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2 Getting Started

This chapter will familiarize you with the framework we shall use throughout the book to think about the design and analysis of algorithms. It is self-contained, but it does include several references to material that we introduce in Chapters 3 and 4. (It also contains several summations, which Appendix A shows how to solve.)

We begin by examining the insertion sort algorithm to solve the sorting problem introduced in Chapter 1. We define a "pseudocode" that should be familiar to you if you have done computer programming, and we use it to show how we shall specify our algorithms. Having specified the insertion sort algorithm, we then argue that it correctly sorts, and we analyze its running time. The analysis introduces a notation that focuses on how that time increases with the number of items to be sorted. Following our discussion of insertion sort, we introduce the divide-and-conquer approach to the design of algorithms and use it to develop an algorithm called merge sort. We end with an analysis of merge sort's running time.

2.1 Insertion sort

Our first algorithm, insertion sort, solves the *sorting problem* introduced in Chapter 1:

- **Input:** A sequence of *n* numbers $\langle a_1, a_2, \ldots, a_n \rangle$.
- **Output:** A permutation (reordering) $\langle a'_1, a'_2, \dots, a'_n \rangle$ of the input sequence such that $a'_1 \leq a'_2 \leq \cdots \leq a'_n$.

The numbers that we wish to sort are also known as the *keys*. Although conceptually we are sorting a sequence, the input comes to us in the form of an array with *n* elements.

In this book, we shall typically describe algorithms as programs written in a *pseudocode* that is similar in many respects to C, C++, Java, Python, or Pascal. If you have been introduced to any of these languages, you should have little trouble



Figure 2.1 Sorting a hand of cards using insertion sort.

reading our algorithms. What separates pseudocode from "real" code is that in pseudocode, we employ whatever expressive method is most clear and concise to specify a given algorithm. Sometimes, the clearest method is English, so do not be surprised if you come across an English phrase or sentence embedded within a section of "real" code. Another difference between pseudocode and real code is that pseudocode is not typically concerned with issues of software engineering. Issues of data abstraction, modularity, and error handling are often ignored in order to convey the essence of the algorithm more concisely.

We start with *insertion sort*, which is an efficient algorithm for sorting a small number of elements. Insertion sort works the way many people sort a hand of playing cards. We start with an empty left hand and the cards face down on the table. We then remove one card at a time from the table and insert it into the correct position in the left hand. To find the correct position for a card, we compare it with each of the cards already in the hand, from right to left, as illustrated in Figure 2.1. At all times, the cards held in the left hand are sorted, and these cards were originally the top cards of the pile on the table.

We present our pseudocode for insertion sort as a procedure called INSERTION-SORT, which takes as a parameter an array A[1..n] containing a sequence of length *n* that is to be sorted. (In the code, the number *n* of elements in *A* is denoted by *A.length.*) The algorithm sorts the input numbers *in place*: it rearranges the numbers within the array *A*, with at most a constant number of them stored outside the array at any time. The input array *A* contains the sorted output sequence when the INSERTION-SORT procedure is finished.



Figure 2.2 The operation of INSERTION-SORT on the array $A = \langle 5, 2, 4, 6, 1, 3 \rangle$. Array indices appear above the rectangles, and values stored in the array positions appear within the rectangles. (a)–(e) The iterations of the **for** loop of lines 1–8. In each iteration, the black rectangle holds the key taken from A[j], which is compared with the values in shaded rectangles to its left in the test of line 5. Shaded arrows show array values moved one position to the right in line 6, and black arrows indicate where the key moves to in line 8. (**f**) The final sorted array.

INSERTION-SORT(A)

1 for j = 2 to A.length 2 kev = A[i]// Insert A[j] into the sorted sequence $A[1 \dots j - 1]$. 3 i = j - 14 5 while i > 0 and A[i] > keyA[i + 1] = A[i]6 i = i - 17 8 A[i+1] = kev

Loop invariants and the correctness of insertion sort

Figure 2.2 shows how this algorithm works for $A = \langle 5, 2, 4, 6, 1, 3 \rangle$. The index *j* indicates the "current card" being inserted into the hand. At the beginning of each iteration of the **for** loop, which is indexed by *j*, the subarray consisting of elements A[1..j-1] constitutes the currently sorted hand, and the remaining subarray A[j+1..n] corresponds to the pile of cards still on the table. In fact, elements A[1..j-1] are the elements *originally* in positions 1 through j-1, but now in sorted order. We state these properties of A[1..j-1] formally as a *loop invariant*:

At the start of each iteration of the **for** loop of lines 1–8, the subarray A[1 ... j - 1] consists of the elements originally in A[1 ... j - 1], but in sorted order.

We use loop invariants to help us understand why an algorithm is correct. We must show three things about a loop invariant:



Initialization: It is true prior to the first iteration of the loop.

- **Maintenance:** If it is true before an iteration of the loop, it remains true before the next iteration.
- **Termination:** When the loop terminates, the invariant gives us a useful property that helps show that the algorithm is correct.

When the first two properties hold, the loop invariant is true prior to every iteration of the loop. (Of course, we are free to use established facts other than the loop invariant itself to prove that the loop invariant remains true before each iteration.) Note the similarity to mathematical induction, where to prove that a property holds, you prove a base case and an inductive step. Here, showing that the invariant holds before the first iteration corresponds to the base case, and showing that the invariant holds from iteration to iteration corresponds to the inductive step.

The third property is perhaps the most important one, since we are using the loop invariant to show correctness. Typically, we use the loop invariant along with the condition that caused the loop to terminate. The termination property differs from how we usually use mathematical induction, in which we apply the inductive step infinitely; here, we stop the "induction" when the loop terminates.

Let us see how these properties hold for insertion sort.

- **Initialization:** We start by showing that the loop invariant holds before the first loop iteration, when j = 2.¹ The subarray A[1 . . j 1], therefore, consists of just the single element A[1], which is in fact the original element in A[1]. Moreover, this subarray is sorted (trivially, of course), which shows that the loop invariant holds prior to the first iteration of the loop.
- **Maintenance:** Next, we tackle the second property: showing that each iteration maintains the loop invariant. Informally, the body of the **for** loop works by moving A[j 1], A[j 2], A[j 3], and so on by one position to the right until it finds the proper position for A[j] (lines 4–7), at which point it inserts the value of A[j] (line 8). The subarray A[1 . . j] then consists of the elements originally in A[1 . . j], but in sorted order. Incrementing j for the next iteration of the **for** loop then preserves the loop invariant.

A more formal treatment of the second property would require us to state and show a loop invariant for the **while** loop of lines 5–7. At this point, however,

¹When the loop is a **for** loop, the moment at which we check the loop invariant just prior to the first iteration is immediately after the initial assignment to the loop-counter variable and just before the first test in the loop header. In the case of INSERTION-SORT, this time is after assigning 2 to the variable *j* but before the first test of whether $j \le A.length$.

we prefer not to get bogged down in such formalism, and so we rely on our informal analysis to show that the second property holds for the outer loop.

Termination: Finally, we examine what happens when the loop terminates. The condition causing the **for** loop to terminate is that j > A.length = n. Because each loop iteration increases j by 1, we must have j = n + 1 at that time. Substituting n + 1 for j in the wording of loop invariant, we have that the subarray A[1..n] consists of the elements originally in A[1..n], but in sorted order. Observing that the subarray A[1..n] is the entire array, we conclude that the entire array is sorted. Hence, the algorithm is correct.

We shall use this method of loop invariants to show correctness later in this chapter and in other chapters as well.

Pseudocode conventions

We use the following conventions in our pseudocode.

- Indentation indicates block structure. For example, the body of the **for** loop that begins on line 1 consists of lines 2–8, and the body of the **while** loop that begins on line 5 contains lines 6–7 but not line 8. Our indentation style applies to **if-else** statements² as well. Using indentation instead of conventional indicators of block structure, such as **begin** and **end** statements, greatly reduces clutter while preserving, or even enhancing, clarity.³
- The looping constructs while, for, and repeat-until and the if-else conditional construct have interpretations similar to those in C, C++, Java, Python, and Pascal.⁴ In this book, the loop counter retains its value after exiting the loop, unlike some situations that arise in C++, Java, and Pascal. Thus, immediately after a for loop, the loop counter's value is the value that first exceeded the for loop bound. We used this property in our correctness argument for insertion sort. The for loop header in line 1 is for j = 2 to A.length, and so when this loop terminates, j = A.length + 1 (or, equivalently, j = n + 1, since n = A.length). We use the keyword to when a for loop increments its loop

⁴Most block-structured languages have equivalent constructs, though the exact syntax may differ. Python lacks **repeat-until** loops, and its **for** loops operate a little differently from the **for** loops in this book.



²In an **if-else** statement, we indent **else** at the same level as its matching **if**. Although we omit the keyword **then**, we occasionally refer to the portion executed when the test following **if** is true as a *then clause*. For multiway tests, we use **elseif** for tests after the first one.

³Each pseudocode procedure in this book appears on one page so that you will not have to discern levels of indentation in code that is split across pages.

counter in each iteration, and we use the keyword **downto** when a **for** loop decrements its loop counter. When the loop counter changes by an amount greater than 1, the amount of change follows the optional keyword **by**.

- The symbol "//" indicates that the remainder of the line is a comment.
- A multiple assignment of the form i = j = e assigns to both variables i and j the value of expression e; it should be treated as equivalent to the assignment j = e followed by the assignment i = j.
- Variables (such as *i*, *j*, and *key*) are local to the given procedure. We shall not use global variables without explicit indication.
- We access array elements by specifying the array name followed by the index in square brackets. For example, A[i] indicates the *i*th element of the array A. The notation "..." is used to indicate a range of values within an array. Thus, A[1...j] indicates the subarray of A consisting of the *j* elements A[1], A[2],..., A[j].
- We typically organize compound data into *objects*, which are composed of *attributes*. We access a particular attribute using the syntax found in many object-oriented programming languages: the object name, followed by a dot, followed by the attribute name. For example, we treat an array as an object with the attribute *length* indicating how many elements it contains. To specify the number of elements in an array *A*, we write *A.length*.

We treat a variable representing an array or object as a pointer to the data representing the array or object. For all attributes f of an object x, setting y = x causes y.f to equal x.f. Moreover, if we now set x.f = 3, then afterward not only does x.f equal 3, but y.f equals 3 as well. In other words, x and y point to the same object after the assignment y = x.

Our attribute notation can "cascade." For example, suppose that the attribute f is itself a pointer to some type of object that has an attribute g. Then the notation x.f.g is implicitly parenthesized as (x.f).g. In other words, if we had assigned y = x.f, then x.f.g is the same as y.g.

Sometimes, a pointer will refer to no object at all. In this case, we give it the special value NIL.

• We pass parameters to a procedure *by value*: the called procedure receives its own copy of the parameters, and if it assigns a value to a parameter, the change is *not* seen by the calling procedure. When objects are passed, the pointer to the data representing the object is copied, but the object's attributes are not. For example, if x is a parameter of a called procedure, the assignment x = y within the called procedure is not visible to the calling procedure. The assignment x.f = 3, however, is visible. Similarly, arrays are passed by pointer, so that

a pointer to the array is passed, rather than the entire array, and changes to individual array elements are visible to the calling procedure.

- A **return** statement immediately transfers control back to the point of call in the calling procedure. Most **return** statements also take a value to pass back to the caller. Our pseudocode differs from many programming languages in that we allow multiple values to be returned in a single **return** statement.
- The boolean operators "and" and "or" are *short circuiting*. That is, when we evaluate the expression "x and y" we first evaluate x. If x evaluates to FALSE, then the entire expression cannot evaluate to TRUE, and so we do not evaluate y. If, on the other hand, x evaluates to TRUE, we must evaluate y to determine the value of the entire expression. Similarly, in the expression "x or y" we evaluate the expression y only if x evaluates to FALSE. Short-circuiting operators allow us to write boolean expressions such as " $x \neq$ NIL and x.f = y" without worrying about what happens when we try to evaluate x.f when x is NIL.
- The keyword **error** indicates that an error occurred because conditions were wrong for the procedure to have been called. The calling procedure is responsible for handling the error, and so we do not specify what action to take.

Exercises

2.1-1

Using Figure 2.2 as a model, illustrate the operation of INSERTION-SORT on the array $A = \langle 31, 41, 59, 26, 41, 58 \rangle$.

2.1-2

Rewrite the INSERTION-SORT procedure to sort into nonincreasing instead of nondecreasing order.

2.1-3

Consider the *searching problem*:

Input: A sequence of *n* numbers $A = \langle a_1, a_2, \dots, a_n \rangle$ and a value ν .

Output: An index *i* such that v = A[i] or the special value NIL if v does not appear in A.

Write pseudocode for *linear search*, which scans through the sequence, looking for ν . Using a loop invariant, prove that your algorithm is correct. Make sure that your loop invariant fulfills the three necessary properties.

2.1-4

Consider the problem of adding two n-bit binary integers, stored in two n-element arrays A and B. The sum of the two integers should be stored in binary form in

an (n + 1)-element array C. State the problem formally and write pseudocode for adding the two integers.

2.2 Analyzing algorithms

Analyzing an algorithm has come to mean predicting the resources that the algorithm requires. Occasionally, resources such as memory, communication bandwidth, or computer hardware are of primary concern, but most often it is computational time that we want to measure. Generally, by analyzing several candidate algorithms for a problem, we can identify a most efficient one. Such analysis may indicate more than one viable candidate, but we can often discard several inferior algorithms in the process.

Before we can analyze an algorithm, we must have a model of the implementation technology that we will use, including a model for the resources of that technology and their costs. For most of this book, we shall assume a generic oneprocessor, *random-access machine (RAM)* model of computation as our implementation technology and understand that our algorithms will be implemented as computer programs. In the RAM model, instructions are executed one after another, with no concurrent operations.

Strictly speaking, we should precisely define the instructions of the RAM model and their costs. To do so, however, would be tedious and would yield little insight into algorithm design and analysis. Yet we must be careful not to abuse the RAM model. For example, what if a RAM had an instruction that sorts? Then we could sort in just one instruction. Such a RAM would be unrealistic, since real computers do not have such instructions. Our guide, therefore, is how real computers are designed. The RAM model contains instructions commonly found in real computers: arithmetic (such as add, subtract, multiply, divide, remainder, floor, ceiling), data movement (load, store, copy), and control (conditional and unconditional branch, subroutine call and return). Each such instruction takes a constant amount of time.

The data types in the RAM model are integer and floating point (for storing real numbers). Although we typically do not concern ourselves with precision in this book, in some applications precision is crucial. We also assume a limit on the size of each word of data. For example, when working with inputs of size n, we typically assume that integers are represented by $c \lg n$ bits for some constant $c \ge 1$. We require $c \ge 1$ so that each word can hold the value of n, enabling us to index the individual input elements, and we restrict c to be a constant so that the word size does not grow arbitrarily. (If the word size could grow arbitrarily, we could store huge amounts of data in one word and operate on it all in constant time—clearly an unrealistic scenario.)

Real computers contain instructions not listed above, and such instructions represent a gray area in the RAM model. For example, is exponentiation a constant-time instruction? In the general case, no; it takes several instructions to compute x^y when x and y are real numbers. In restricted situations, however, exponentiation is a constant-time operation. Many computers have a "shift left" instruction, which in constant time shifts the bits of an integer by k positions to the left. In most computers, shifting the bits of an integer by one position to the left is equivalent to multiplication by 2, so that shifting the bits by k positions to the left is equivalent to multiplication by 2^k . Therefore, such computers can compute 2^k in one constant-time instruction by shifting the integer 1 by k positions to the left, as long as k is no more than the number of bits in a computer word. We will endeavor to avoid such gray areas in the RAM model, but we will treat computation of 2^k as a constant-time operation when k is a small enough positive integer.

In the RAM model, we do not attempt to model the memory hierarchy that is common in contemporary computers. That is, we do not model caches or virtual memory. Several computational models attempt to account for memory-hierarchy effects, which are sometimes significant in real programs on real machines. A handful of problems in this book examine memory-hierarchy effects, but for the most part, the analyses in this book will not consider them. Models that include the memory hierarchy are quite a bit more complex than the RAM model, and so they can be difficult to work with. Moreover, RAM-model analyses are usually excellent predictors of performance on actual machines.

Analyzing even a simple algorithm in the RAM model can be a challenge. The mathematical tools required may include combinatorics, probability theory, algebraic dexterity, and the ability to identify the most significant terms in a formula. Because the behavior of an algorithm may be different for each possible input, we need a means for summarizing that behavior in simple, easily understood formulas.

Even though we typically select only one machine model to analyze a given algorithm, we still face many choices in deciding how to express our analysis. We would like a way that is simple to write and manipulate, shows the important characteristics of an algorithm's resource requirements, and suppresses tedious details.

Analysis of insertion sort

The time taken by the INSERTION-SORT procedure depends on the input: sorting a thousand numbers takes longer than sorting three numbers. Moreover, INSERTION-SORT can take different amounts of time to sort two input sequences of the same size depending on how nearly sorted they already are. In general, the time taken by an algorithm grows with the size of the input, so it is traditional to describe the running time of a program as a function of the size of its input. To do so, we need to define the terms "running time" and "size of input" more carefully.

The best notion for *input size* depends on the problem being studied. For many problems, such as sorting or computing discrete Fourier transforms, the most natural measure is the *number of items in the input*—for example, the array size n for sorting. For many other problems, such as multiplying two integers, the best measure of input size is the *total number of bits* needed to represent the input in ordinary binary notation. Sometimes, it is more appropriate to describe the size of the input with two numbers rather than one. For instance, if the input to an algorithm is a graph, the input size can be described by the numbers of vertices and edges in the graph. We shall indicate which input size measure is being used with each problem we study.

The *running time* of an algorithm on a particular input is the number of primitive operations or "steps" executed. It is convenient to define the notion of step so that it is as machine-independent as possible. For the moment, let us adopt the following view. A constant amount of time is required to execute each line of our pseudocode. One line may take a different amount of time than another line, but we shall assume that each execution of the *i*th line takes time c_i , where c_i is a constant. This viewpoint is in keeping with the RAM model, and it also reflects how the pseudocode would be implemented on most actual computers.⁵

In the following discussion, our expression for the running time of INSERTION-SORT will evolve from a messy formula that uses all the statement costs c_i to a much simpler notation that is more concise and more easily manipulated. This simpler notation will also make it easy to determine whether one algorithm is more efficient than another.

We start by presenting the INSERTION-SORT procedure with the time "cost" of each statement and the number of times each statement is executed. For each j = 2, 3, ..., n, where n = A.length, we let t_j denote the number of times the **while** loop test in line 5 is executed for that value of j. When a **for** or **while** loop exits in the usual way (i.e., due to the test in the loop header), the test is executed one time more than the loop body. We assume that comments are not executable statements, and so they take no time.

⁵There are some subtleties here. Computational steps that we specify in English are often variants of a procedure that requires more than just a constant amount of time. For example, later in this book we might say "sort the points by *x*-coordinate," which, as we shall see, takes more than a constant amount of time. Also, note that a statement that calls a subroutine takes constant time, though the subroutine, once invoked, may take more. That is, we separate the process of *calling* the subroutine – passing parameters to it, etc. – from the process of *executing* the subroutine.

INSERTION-SORT (A)		cost	times
1	for $j = 2$ to A.length	c_1	п
2	key = A[j]	<i>C</i> ₂	n - 1
3	// Insert $A[j]$ into the sorted		
	sequence $A[1 \dots j - 1]$.	0	n - 1
4	i = j - 1	C4	n - 1
5	while $i > 0$ and $A[i] > key$	C ₅	$\sum_{j=2}^{n} t_j$
6	A[i+1] = A[i]	<i>C</i> ₆	$\sum_{j=2}^{n} (t_j - 1)$
7	i = i - 1	C7	$\sum_{j=2}^{n} (t_j - 1)$
8	A[i+1] = key	C 8	n-1

The running time of the algorithm is the sum of running times for each statement executed; a statement that takes c_i steps to execute and executes n times will contribute $c_i n$ to the total running time.⁶ To compute T(n), the running time of INSERTION-SORT on an input of n values, we sum the products of the *cost* and *times* columns, obtaining

$$T(n) = c_1 n + c_2 (n-1) + c_4 (n-1) + c_5 \sum_{j=2}^n t_j + c_6 \sum_{j=2}^n (t_j - 1) + c_7 \sum_{j=2}^n (t_j - 1) + c_8 (n-1).$$

Even for inputs of a given size, an algorithm's running time may depend on *which* input of that size is given. For example, in INSERTION-SORT, the best case occurs if the array is already sorted. For each j = 2, 3, ..., n, we then find that $A[i] \leq key$ in line 5 when *i* has its initial value of j - 1. Thus $t_j = 1$ for j = 2, 3, ..., n, and the best-case running time is

$$T(n) = c_1 n + c_2(n-1) + c_4(n-1) + c_5(n-1) + c_8(n-1)$$

= $(c_1 + c_2 + c_4 + c_5 + c_8)n - (c_2 + c_4 + c_5 + c_8)$.

We can express this running time as an + b for *constants* a and b that depend on the statement costs c_i ; it is thus a *linear function* of n.

If the array is in reverse sorted order—that is, in decreasing order—the worst case results. We must compare each element A[j] with each element in the entire sorted subarray A[1..j-1], and so $t_j = j$ for j = 2, 3, ..., n. Noting that

⁶This characteristic does not necessarily hold for a resource such as memory. A statement that references m words of memory and is executed n times does not necessarily reference mn distinct words of memory.



$$\sum_{j=2}^{n} j = \frac{n(n+1)}{2} - 1$$

and

$$\sum_{j=2}^{n} (j-1) = \frac{n(n-1)}{2}$$

(see Appendix A for a review of how to solve these summations), we find that in the worst case, the running time of INSERTION-SORT is

$$T(n) = c_1 n + c_2 (n-1) + c_4 (n-1) + c_5 \left(\frac{n(n+1)}{2} - 1\right) + c_6 \left(\frac{n(n-1)}{2}\right) + c_7 \left(\frac{n(n-1)}{2}\right) + c_8 (n-1) = \left(\frac{c_5}{2} + \frac{c_6}{2} + \frac{c_7}{2}\right) n^2 + \left(c_1 + c_2 + c_4 + \frac{c_5}{2} - \frac{c_6}{2} - \frac{c_7}{2} + c_8\right) n - (c_2 + c_4 + c_5 + c_8).$$

We can express this worst-case running time as $an^2 + bn + c$ for constants a, b, and c that again depend on the statement costs c_i ; it is thus a *quadratic function* of n.

Typically, as in insertion sort, the running time of an algorithm is fixed for a given input, although in later chapters we shall see some interesting "randomized" algorithms whose behavior can vary even for a fixed input.

Worst-case and average-case analysis

In our analysis of insertion sort, we looked at both the best case, in which the input array was already sorted, and the worst case, in which the input array was reverse sorted. For the remainder of this book, though, we shall usually concentrate on finding only the *worst-case running time*, that is, the longest running time for *any* input of size n. We give three reasons for this orientation.

- The worst-case running time of an algorithm gives us an upper bound on the running time for any input. Knowing it provides a guarantee that the algorithm will never take any longer. We need not make some educated guess about the running time and hope that it never gets much worse.
- For some algorithms, the worst case occurs fairly often. For example, in searching a database for a particular piece of information, the searching algorithm's worst case will often occur when the information is not present in the database. In some applications, searches for absent information may be frequent.

The "average case" is often roughly as bad as the worst case. Suppose that we randomly choose *n* numbers and apply insertion sort. How long does it take to determine where in subarray A[1 ... j - 1] to insert element A[j]? On average, half the elements in A[1 ... j - 1] are less than A[j], and half the elements are greater. On average, therefore, we check half of the subarray A[1 ... j - 1], and so t_j is about j/2. The resulting average-case running time turns out to be a quadratic function of the input size, just like the worst-case running time.

In some particular cases, we shall be interested in the *average-case* running time of an algorithm; we shall see the technique of *probabilistic analysis* applied to various algorithms throughout this book. The scope of average-case analysis is limited, because it may not be apparent what constitutes an "average" input for a particular problem. Often, we shall assume that all inputs of a given size are equally likely. In practice, this assumption may be violated, but we can sometimes use a *randomized algorithm*, which makes random choices, to allow a probabilistic analysis and yield an *expected* running time. We explore randomized algorithms more in Chapter 5 and in several other subsequent chapters.

Order of growth

We used some simplifying abstractions to ease our analysis of the INSERTION-SORT procedure. First, we ignored the actual cost of each statement, using the constants c_i to represent these costs. Then, we observed that even these constants give us more detail than we really need: we expressed the worst-case running time as $an^2 + bn + c$ for some constants a, b, and c that depend on the statement costs c_i . We thus ignored not only the actual statement costs, but also the abstract costs c_i .

We shall now make one more simplifying abstraction: it is the *rate of growth*, or *order of growth*, of the running time that really interests us. We therefore consider only the leading term of a formula (e.g., an^2), since the lower-order terms are relatively insignificant for large values of n. We also ignore the leading term's constant coefficient, since constant factors are less significant than the rate of growth in determining computational efficiency for large inputs. For insertion sort, when we ignore the lower-order terms and the leading term's constant coefficient, we are left with the factor of n^2 from the leading term. We write that insertion sort has a worst-case running time of $\Theta(n^2)$ (pronounced "theta of n-squared"). We shall use Θ -notation informally in this chapter, and we will define it precisely in Chapter 3.

We usually consider one algorithm to be more efficient than another if its worstcase running time has a lower order of growth. Due to constant factors and lowerorder terms, an algorithm whose running time has a higher order of growth might take less time for small inputs than an algorithm whose running time has a lower

order of growth. But for large enough inputs, a $\Theta(n^2)$ algorithm, for example, will run more quickly in the worst case than a $\Theta(n^3)$ algorithm.

Exercises

2.2-1

Express the function $n^3/1000 - 100n^2 - 100n + 3$ in terms of Θ -notation.

2.2-2

Consider sorting *n* numbers stored in array *A* by first finding the smallest element of *A* and exchanging it with the element in *A*[1]. Then find the second smallest element of *A*, and exchange it with *A*[2]. Continue in this manner for the first n - 1elements of *A*. Write pseudocode for this algorithm, which is known as *selection sort*. What loop invariant does this algorithm maintain? Why does it need to run for only the first n - 1 elements, rather than for all *n* elements? Give the best-case and worst-case running times of selection sort in Θ -notation.

2.2-3

Consider linear search again (see Exercise 2.1-3). How many elements of the input sequence need to be checked on the average, assuming that the element being searched for is equally likely to be any element in the array? How about in the worst case? What are the average-case and worst-case running times of linear search in Θ -notation? Justify your answers.

2.2-4

How can we modify almost any algorithm to have a good best-case running time?

2.3 Designing algorithms

We can choose from a wide range of algorithm design techniques. For insertion sort, we used an *incremental* approach: having sorted the subarray A[1.. j - 1], we inserted the single element A[j] into its proper place, yielding the sorted subarray A[1.. j].

In this section, we examine an alternative design approach, known as "divideand-conquer," which we shall explore in more detail in Chapter 4. We'll use divideand-conquer to design a sorting algorithm whose worst-case running time is much less than that of insertion sort. One advantage of divide-and-conquer algorithms is that their running times are often easily determined using techniques that we will see in Chapter 4.

2.3.1 The divide-and-conquer approach

Many useful algorithms are *recursive* in structure: to solve a given problem, they call themselves recursively one or more times to deal with closely related subproblems. These algorithms typically follow a *divide-and-conquer* approach: they break the problem into several subproblems that are similar to the original problem but smaller in size, solve the subproblems recursively, and then combine these solutions to create a solution to the original problem.

The divide-and-conquer paradigm involves three steps at each level of the recursion:

- **Divide** the problem into a number of subproblems that are smaller instances of the same problem.
- **Conquer** the subproblems by solving them recursively. If the subproblem sizes are small enough, however, just solve the subproblems in a straightforward manner.
- **Combine** the solutions to the subproblems into the solution for the original problem.

The *merge sort* algorithm closely follows the divide-and-conquer paradigm. Intuitively, it operates as follows.

- **Divide:** Divide the *n*-element sequence to be sorted into two subsequences of n/2 elements each.
- **Conquer:** Sort the two subsequences recursively using merge sort.

Combine: Merge the two sorted subsequences to produce the sorted answer.

The recursion "bottoms out" when the sequence to be sorted has length 1, in which case there is no work to be done, since every sequence of length 1 is already in sorted order.

The key operation of the merge sort algorithm is the merging of two sorted sequences in the "combine" step. We merge by calling an auxiliary procedure MERGE(A, p, q, r), where A is an array and p, q, and r are indices into the array such that $p \le q < r$. The procedure assumes that the subarrays $A[p \dots q]$ and $A[q+1 \dots r]$ are in sorted order. It **merges** them to form a single sorted subarray that replaces the current subarray $A[p \dots r]$.

Our MERGE procedure takes time $\Theta(n)$, where n = r - p + 1 is the total number of elements being merged, and it works as follows. Returning to our cardplaying motif, suppose we have two piles of cards face up on a table. Each pile is sorted, with the smallest cards on top. We wish to merge the two piles into a single sorted output pile, which is to be face down on the table. Our basic step consists of choosing the smaller of the two cards on top of the face-up piles, removing it from its pile (which exposes a new top card), and placing this card face down onto

the output pile. We repeat this step until one input pile is empty, at which time we just take the remaining input pile and place it face down onto the output pile. Computationally, each basic step takes constant time, since we are comparing just the two top cards. Since we perform at most n basic steps, merging takes $\Theta(n)$ time.

The following pseudocode implements the above idea, but with an additional twist that avoids having to check whether either pile is empty in each basic step. We place on the bottom of each pile a *sentinel* card, which contains a special value that we use to simplify our code. Here, we use ∞ as the sentinel value, so that whenever a card with ∞ is exposed, it cannot be the smaller card unless both piles have their sentinel cards exposed. But once that happens, all the nonsentinel cards have already been placed onto the output pile. Since we know in advance that exactly r - p + 1 cards will be placed onto the output pile, we can stop once we have performed that many basic steps.

MERGE(A, p, q, r)

 $1 \quad n_1 = q - p + 1$ 2 $n_2 = r - q$ 3 let $L[1 \dots n_1 + 1]$ and $R[1 \dots n_2 + 1]$ be new arrays 4 for i = 1 to n_1 5 L[i] = A[p+i-1]6 **for** j = 1 **to** n_2 R[j] = A[q+j]7 $L[n_1+1] = \infty$ 8 9 $R[n_2+1] = \infty$ 10 i = 111 i = 112 for k = p to rif $L[i] \leq R[j]$ 13 14 A[k] = L[i]i = i + 115 else A[k] = R[j]16 17 i = i + 1

In detail, the MERGE procedure works as follows. Line 1 computes the length n_1 of the subarray A[p ...q], and line 2 computes the length n_2 of the subarray A[q + 1...r]. We create arrays L and R ("left" and "right"), of lengths $n_1 + 1$ and $n_2 + 1$, respectively, in line 3; the extra position in each array will hold the sentinel. The **for** loop of lines 4–5 copies the subarray A[p ...q] into $L[1...n_1]$, and the **for** loop of lines 6–7 copies the subarray A[q + 1...r] into $R[1...n_2]$. Lines 8–9 put the sentinels at the ends of the arrays L and R. Lines 10–17, illus-



Figure 2.3 The operation of lines 10–17 in the call MERGE(A, 9, 12, 16), when the subarray A[9..16] contains the sequence $\langle 2, 4, 5, 7, 1, 2, 3, 6 \rangle$. After copying and inserting sentinels, the array L contains $\langle 2, 4, 5, 7, \infty \rangle$, and the array R contains $\langle 1, 2, 3, 6, \infty \rangle$. Lightly shaded positions in A contain their final values, and lightly shaded positions in L and R contain values that have yet to be copied back into A. Taken together, the lightly shaded positions always comprise the values originally in A[9..16], along with the two sentinels. Heavily shaded positions in A contain values that have already been copied back into A. (a)–(h) The arrays A, L, and R, and their respective indices k, i, and j prior to each iteration of the loop of lines 12–17.

trated in Figure 2.3, perform the r - p + 1 basic steps by maintaining the following loop invariant:

At the start of each iteration of the **for** loop of lines 12–17, the subarray A[p..k-1] contains the k - p smallest elements of $L[1..n_1 + 1]$ and $R[1..n_2 + 1]$, in sorted order. Moreover, L[i] and R[j] are the smallest elements of their arrays that have not been copied back into A.

We must show that this loop invariant holds prior to the first iteration of the **for** loop of lines 12–17, that each iteration of the loop maintains the invariant, and that the invariant provides a useful property to show correctness when the loop terminates.

Initialization: Prior to the first iteration of the loop, we have k = p, so that the subarray $A[p \dots k - 1]$ is empty. This empty subarray contains the k - p = 0 smallest elements of L and R, and since i = j = 1, both L[i] and R[j] are the smallest elements of their arrays that have not been copied back into A.



Figure 2.3, continued (i) The arrays and indices at termination. At this point, the subarray in A[9..16] is sorted, and the two sentinels in L and R are the only two elements in these arrays that have not been copied into A.

- **Maintenance:** To see that each iteration maintains the loop invariant, let us first suppose that $L[i] \leq R[j]$. Then L[i] is the smallest element not yet copied back into A. Because $A[p \dots k 1]$ contains the k p smallest elements, after line 14 copies L[i] into A[k], the subarray $A[p \dots k]$ will contain the k p + 1 smallest elements. Incrementing k (in the **for** loop update) and i (in line 15) reestablishes the loop invariant for the next iteration. If instead L[i] > R[j], then lines 16–17 perform the appropriate action to maintain the loop invariant.
- **Termination:** At termination, k = r + 1. By the loop invariant, the subarray A[p ... k 1], which is A[p ... r], contains the k p = r p + 1 smallest elements of $L[1... n_1 + 1]$ and $R[1... n_2 + 1]$, in sorted order. The arrays L and R together contain $n_1 + n_2 + 2 = r p + 3$ elements. All but the two largest have been copied back into A, and these two largest elements are the sentinels.

To see that the MERGE procedure runs in $\Theta(n)$ time, where n = r - p + 1, observe that each of lines 1–3 and 8–11 takes constant time, the **for** loops of lines 4–7 take $\Theta(n_1 + n_2) = \Theta(n)$ time,⁷ and there are *n* iterations of the **for** loop of lines 12–17, each of which takes constant time.

We can now use the MERGE procedure as a subroutine in the merge sort algorithm. The procedure MERGE-SORT(A, p, r) sorts the elements in the subarray A[p ...r]. If $p \ge r$, the subarray has at most one element and is therefore already sorted. Otherwise, the divide step simply computes an index q that partitions A[p ...r] into two subarrays: A[p ...q], containing $\lceil n/2 \rceil$ elements, and A[q + 1...r], containing $\lceil n/2 \rceil$ elements.⁸

MERGE-SORT(A, p, r)

1 **if** p < r2 $q = \lfloor (p+r)/2 \rfloor$ 3 MERGE-SORT(A, p, q)4 MERGE-SORT(A, q+1, r)5 MERGE(A, p, q, r)

To sort the entire sequence $A = \langle A[1], A[2], \ldots, A[n] \rangle$, we make the initial call MERGE-SORT(A, 1, A.length), where once again A.length = n. Figure 2.4 illustrates the operation of the procedure bottom-up when n is a power of 2. The algorithm consists of merging pairs of 1-item sequences to form sorted sequences of length 2, merging pairs of sequences of length 2 to form sorted sequences of length 4, and so on, until two sequences of length n/2 are merged to form the final sorted sequence of length n.

2.3.2 Analyzing divide-and-conquer algorithms

When an algorithm contains a recursive call to itself, we can often describe its running time by a *recurrence equation* or *recurrence*, which describes the overall running time on a problem of size *n* in terms of the running time on smaller inputs. We can then use mathematical tools to solve the recurrence and provide bounds on the performance of the algorithm.

⁸The expression $\lceil x \rceil$ denotes the least integer greater than or equal to x, and $\lfloor x \rfloor$ denotes the greatest integer less than or equal to x. These notations are defined in Chapter 3. The easiest way to verify that setting q to $\lfloor (p+r)/2 \rfloor$ yields subarrays A[p ...q] and A[q+1...r] of sizes $\lceil n/2 \rceil$ and $\lfloor n/2 \rfloor$, respectively, is to examine the four cases that arise depending on whether each of p and r is odd or even.



⁷We shall see in Chapter 3 how to formally interpret equations containing Θ -notation.



Figure 2.4 The operation of merge sort on the array A = (5, 2, 4, 7, 1, 3, 2, 6). The lengths of the sorted sequences being merged increase as the algorithm progresses from bottom to top.

A recurrence for the running time of a divide-and-conquer algorithm falls out from the three steps of the basic paradigm. As before, we let T(n) be the running time on a problem of size n. If the problem size is small enough, say $n \le c$ for some constant c, the straightforward solution takes constant time, which we write as $\Theta(1)$. Suppose that our division of the problem yields a subproblems, each of which is 1/b the size of the original. (For merge sort, both a and b are 2, but we shall see many divide-and-conquer algorithms in which $a \ne b$.) It takes time T(n/b) to solve one subproblem of size n/b, and so it takes time aT(n/b)to solve a of them. If we take D(n) time to divide the problem into subproblems and C(n) time to combine the solutions to the subproblems into the solution to the original problem, we get the recurrence

$$T(n) = \begin{cases} \Theta(1) & \text{if } n \le c ,\\ aT(n/b) + D(n) + C(n) & \text{otherwise} \end{cases}$$

In Chapter 4, we shall see how to solve common recurrences of this form.

Analysis of merge sort

Although the pseudocode for MERGE-SORT works correctly when the number of elements is not even, our recurrence-based analysis is simplified if we assume that

the original problem size is a power of 2. Each divide step then yields two subsequences of size exactly n/2. In Chapter 4, we shall see that this assumption does not affect the order of growth of the solution to the recurrence.

We reason as follows to set up the recurrence for T(n), the worst-case running time of merge sort on *n* numbers. Merge sort on just one element takes constant time. When we have n > 1 elements, we break down the running time as follows.

- **Divide:** The divide step just computes the middle of the subarray, which takes constant time. Thus, $D(n) = \Theta(1)$.
- **Conquer:** We recursively solve two subproblems, each of size n/2, which contributes 2T(n/2) to the running time.
- **Combine:** We have already noted that the MERGE procedure on an *n*-element subarray takes time $\Theta(n)$, and so $C(n) = \Theta(n)$.

When we add the functions D(n) and C(n) for the merge sort analysis, we are adding a function that is $\Theta(n)$ and a function that is $\Theta(1)$. This sum is a linear function of n, that is, $\Theta(n)$. Adding it to the 2T(n/2) term from the "conquer" step gives the recurrence for the worst-case running time T(n) of merge sort:

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ 2T(n/2) + \Theta(n) & \text{if } n > 1 . \end{cases}$$
(2.1)

In Chapter 4, we shall see the "master theorem," which we can use to show that T(n) is $\Theta(n \lg n)$, where $\lg n$ stands for $\log_2 n$. Because the logarithm function grows more slowly than any linear function, for large enough inputs, merge sort, with its $\Theta(n \lg n)$ running time, outperforms insertion sort, whose running time is $\Theta(n^2)$, in the worst case.

We do not need the master theorem to intuitively understand why the solution to the recurrence (2.1) is $T(n) = \Theta(n \lg n)$. Let us rewrite recurrence (2.1) as

$$T(n) = \begin{cases} c & \text{if } n = 1, \\ 2T(n/2) + cn & \text{if } n > 1, \end{cases}$$
(2.2)

where the constant c represents the time required to solve problems of size 1 as well as the time per array element of the divide and combine steps.⁹

⁹It is unlikely that the same constant exactly represents both the time to solve problems of size 1 and the time per array element of the divide and combine steps. We can get around this problem by letting *c* be the larger of these times and understanding that our recurrence gives an upper bound on the running time, or by letting *c* be the lesser of these times and understanding that our recurrence gives a lower bound on the running time. Both bounds are on the order of $n \lg n$ and, taken together, give a $\Theta(n \lg n)$ running time.

Figure 2.5 shows how we can solve recurrence (2.2). For convenience, we assume that n is an exact power of 2. Part (a) of the figure shows T(n), which we expand in part (b) into an equivalent tree representing the recurrence. The cn term is the root (the cost incurred at the top level of recursion), and the two subtrees of the root are the two smaller recurrences T(n/2). Part (c) shows this process carried one step further by expanding T(n/2). The cost incurred at each of the two subnodes at the second level of recursion is cn/2. We continue expanding each node in the tree by breaking it into its constituent parts as determined by the recurrence, until the problem sizes get down to 1, each with a cost of c. Part (d) shows the resulting *recursion tree*.

Next, we add the costs across each level of the tree. The top level has total cost cn, the next level down has total cost c(n/2) + c(n/2) = cn, the level after that has total cost c(n/4) + c(n/4) + c(n/4) + c(n/4) = cn, and so on. In general, the level *i* below the top has 2^i nodes, each contributing a cost of $c(n/2^i)$, so that the *i*th level below the top has total cost $2^i c(n/2^i) = cn$. The bottom level has *n* nodes, each contributing a cost of *c*.

The total number of levels of the recursion tree in Figure 2.5 is $\lg n + 1$, where n is the number of leaves, corresponding to the input size. An informal inductive argument justifies this claim. The base case occurs when n = 1, in which case the tree has only one level. Since $\lg 1 = 0$, we have that $\lg n + 1$ gives the correct number of levels. Now assume as an inductive hypothesis that the number of levels of a recursion tree with 2^i leaves is $\lg 2^i + 1 = i + 1$ (since for any value of i, we have that $\lg 2^i = i$). Because we are assuming that the input size is a power of 2, the next input size to consider is 2^{i+1} . A tree with $n = 2^{i+1}$ leaves has one more level than a tree with 2^i leaves, and so the total number of levels is $(i + 1) + 1 = \lg 2^{i+1} + 1$.

To compute the total cost represented by the recurrence (2.2), we simply add up the costs of all the levels. The recursion tree has $\lg n + 1$ levels, each costing cn, for a total cost of $cn(\lg n + 1) = cn \lg n + cn$. Ignoring the low-order term and the constant c gives the desired result of $\Theta(n \lg n)$.

Exercises

2.3-1

Using Figure 2.4 as a model, illustrate the operation of merge sort on the array $A = \langle 3, 41, 52, 26, 38, 57, 9, 49 \rangle$.

2.3-2

Rewrite the MERGE procedure so that it does not use sentinels, instead stopping once either array L or R has had all its elements copied back to A and then copying the remainder of the other array back into A.



Figure 2.5 How to construct a recursion tree for the recurrence T(n) = 2T(n/2) + cn. Part (a) shows T(n), which progressively expands in (b)–(d) to form the recursion tree. The fully expanded tree in part (d) has $\lg n + 1$ levels (i.e., it has height $\lg n$, as indicated), and each level contributes a total cost of cn. The total cost, therefore, is $cn \lg n + cn$, which is $\Theta(n \lg n)$.

2.3-3

Use mathematical induction to show that when n is an exact power of 2, the solution of the recurrence

$$T(n) = \begin{cases} 2 & \text{if } n = 2, \\ 2T(n/2) + n & \text{if } n = 2^k, \text{ for } k > 1 \end{cases}$$

is $T(n) = n \lg n.$

2.3-4

We can express insertion sort as a recursive procedure as follows. In order to sort A[1..n], we recursively sort A[1..n-1] and then insert A[n] into the sorted array A[1..n-1]. Write a recurrence for the running time of this recursive version of insertion sort.

2.3-5

Referring back to the searching problem (see Exercise 2.1-3), observe that if the sequence A is sorted, we can check the midpoint of the sequence against v and eliminate half of the sequence from further consideration. The **binary search** algorithm repeats this procedure, halving the size of the remaining portion of the sequence each time. Write pseudocode, either iterative or recursive, for binary search. Argue that the worst-case running time of binary search is $\Theta(\lg n)$.

2.3-6

Observe that the **while** loop of lines 5–7 of the INSERTION-SORT procedure in Section 2.1 uses a linear search to scan (backward) through the sorted subarray A[1 ... j - 1]. Can we use a binary search (see Exercise 2.3-5) instead to improve the overall worst-case running time of insertion sort to $\Theta(n \lg n)$?

2.3-7 *

Describe a $\Theta(n \lg n)$ -time algorithm that, given a set *S* of *n* integers and another integer *x*, determines whether or not there exist two elements in *S* whose sum is exactly *x*.

Problems

2-1 Insertion sort on small arrays in merge sort

Although merge sort runs in $\Theta(n \lg n)$ worst-case time and insertion sort runs in $\Theta(n^2)$ worst-case time, the constant factors in insertion sort can make it faster in practice for small problem sizes on many machines. Thus, it makes sense to *coarsen* the leaves of the recursion by using insertion sort within merge sort when

40

subproblems become sufficiently small. Consider a modification to merge sort in which n/k sublists of length k are sorted using insertion sort and then merged using the standard merging mechanism, where k is a value to be determined.

- *a.* Show that insertion sort can sort the n/k sublists, each of length k, in $\Theta(nk)$ worst-case time.
- **b.** Show how to merge the sublists in $\Theta(n \lg(n/k))$ worst-case time.
- *c.* Given that the modified algorithm runs in $\Theta(nk + n \lg(n/k))$ worst-case time, what is the largest value of *k* as a function of *n* for which the modified algorithm has the same running time as standard merge sort, in terms of Θ -notation?
- *d.* How should we choose *k* in practice?

2-2 Correctness of bubblesort

Bubblesort is a popular, but inefficient, sorting algorithm. It works by repeatedly swapping adjacent elements that are out of order.

BUBBLESORT(A)

1 for i = 1 to A.length -12 for j = A.length downto i + 13 if A[j] < A[j-1]4 exchange A[j] with A[j-1]

a. Let A' denote the output of BUBBLESORT(A). To prove that BUBBLESORT is correct, we need to prove that it terminates and that

$$A'[1] \le A'[2] \le \dots \le A'[n],$$
 (2.3)

where n = A.length. In order to show that BUBBLESORT actually sorts, what else do we need to prove?

The next two parts will prove inequality (2.3).

- **b.** State precisely a loop invariant for the **for** loop in lines 2–4, and prove that this loop invariant holds. Your proof should use the structure of the loop invariant proof presented in this chapter.
- *c.* Using the termination condition of the loop invariant proved in part (b), state a loop invariant for the **for** loop in lines 1–4 that will allow you to prove inequality (2.3). Your proof should use the structure of the loop invariant proof presented in this chapter.

d. What is the worst-case running time of bubblesort? How does it compare to the running time of insertion sort?

2-3 Correctness of Horner's rule

The following code fragment implements Horner's rule for evaluating a polynomial

$$P(x) = \sum_{k=0}^{n} a_k x^k$$

= $a_0 + x(a_1 + x(a_2 + \dots + x(a_{n-1} + xa_n) \dots)),$

given the coefficients a_0, a_1, \ldots, a_n and a value for x:

1
$$y = 0$$

2 for $i = n$ downto 0
3 $y = a_i + x \cdot y$

- *a.* In terms of Θ -notation, what is the running time of this code fragment for Horner's rule?
- **b.** Write pseudocode to implement the naive polynomial-evaluation algorithm that computes each term of the polynomial from scratch. What is the running time of this algorithm? How does it compare to Horner's rule?
- *c*. Consider the following loop invariant:

At the start of each iteration of the **for** loop of lines 2–3,

$$y = \sum_{k=0}^{n-(i+1)} a_{k+i+1} x^k$$
.

Interpret a summation with no terms as equaling 0. Following the structure of the loop invariant proof presented in this chapter, use this loop invariant to show that, at termination, $y = \sum_{k=0}^{n} a_k x^k$.

d. Conclude by arguing that the given code fragment correctly evaluates a polynomial characterized by the coefficients a_0, a_1, \ldots, a_n .

2-4 Inversions

Let A[1..n] be an array of *n* distinct numbers. If i < j and A[i] > A[j], then the pair (i, j) is called an *inversion* of *A*.

a. List the five inversions of the array $\langle 2, 3, 8, 6, 1 \rangle$.

- **b.** What array with elements from the set {1, 2, ..., *n*} has the most inversions? How many does it have?
- *c.* What is the relationship between the running time of insertion sort and the number of inversions in the input array? Justify your answer.
- *d.* Give an algorithm that determines the number of inversions in any permutation on *n* elements in $\Theta(n \lg n)$ worst-case time. (*Hint:* Modify merge sort.)

Chapter notes

In 1968, Knuth published the first of three volumes with the general title *The Art of Computer Programming* [209, 210, 211]. The first volume ushered in the modern study of computer algorithms with a focus on the analysis of running time, and the full series remains an engaging and worthwhile reference for many of the topics presented here. According to Knuth, the word "algorithm" is derived from the name "al-Khowârizmî," a ninth-century Persian mathematician.

Aho, Hopcroft, and Ullman [5] advocated the asymptotic analysis of algorithms—using notations that Chapter 3 introduces, including Θ -notation—as a means of comparing relative performance. They also popularized the use of recurrence relations to describe the running times of recursive algorithms.

Knuth [211] provides an encyclopedic treatment of many sorting algorithms. His comparison of sorting algorithms (page 381) includes exact step-counting analyses, like the one we performed here for insertion sort. Knuth's discussion of insertion sort encompasses several variations of the algorithm. The most important of these is Shell's sort, introduced by D. L. Shell, which uses insertion sort on periodic subsequences of the input to produce a faster sorting algorithm.

Merge sort is also described by Knuth. He mentions that a mechanical collator capable of merging two decks of punched cards in a single pass was invented in 1938. J. von Neumann, one of the pioneers of computer science, apparently wrote a program for merge sort on the EDVAC computer in 1945.

The early history of proving programs correct is described by Gries [153], who credits P. Naur with the first article in this field. Gries attributes loop invariants to R. W. Floyd. The textbook by Mitchell [256] describes more recent progress in proving programs correct.

3 Growth of Functions

The order of growth of the running time of an algorithm, defined in Chapter 2, gives a simple characterization of the algorithm's efficiency and also allows us to compare the relative performance of alternative algorithms. Once the input size n becomes large enough, merge sort, with its $\Theta(n \lg n)$ worst-case running time, beats insertion sort, whose worst-case running time is $\Theta(n^2)$. Although we can sometimes determine the exact running time of an algorithm, as we did for insertion sort in Chapter 2, the extra precision is not usually worth the effort of computing it. For large enough inputs, the multiplicative constants and lower-order terms of an exact running time are dominated by the effects of the input size itself.

When we look at input sizes large enough to make only the order of growth of the running time relevant, we are studying the *asymptotic* efficiency of algorithms. That is, we are concerned with how the running time of an algorithm increases with the size of the input *in the limit*, as the size of the input increases without bound. Usually, an algorithm that is asymptotically more efficient will be the best choice for all but very small inputs.

This chapter gives several standard methods for simplifying the asymptotic analysis of algorithms. The next section begins by defining several types of "asymptotic notation," of which we have already seen an example in Θ -notation. We then present several notational conventions used throughout this book, and finally we review the behavior of functions that commonly arise in the analysis of algorithms.

3.1 Asymptotic notation

The notations we use to describe the asymptotic running time of an algorithm are defined in terms of functions whose domains are the set of natural numbers $\mathbb{N} = \{0, 1, 2, \ldots\}$. Such notations are convenient for describing the worst-case running-time function T(n), which usually is defined only on integer input sizes. We sometimes find it convenient, however, to *abuse* asymptotic notation in a va-

riety of ways. For example, we might extend the notation to the domain of real numbers or, alternatively, restrict it to a subset of the natural numbers. We should make sure, however, to understand the precise meaning of the notation so that when we abuse, we do not *misuse* it. This section defines the basic asymptotic notations and also introduces some common abuses.

Asymptotic notation, functions, and running times

We will use asymptotic notation primarily to describe the running times of algorithms, as when we wrote that insertion sort's worst-case running time is $\Theta(n^2)$. Asymptotic notation actually applies to functions, however. Recall that we characterized insertion sort's worst-case running time as $an^2 + bn + c$, for some constants a, b, and c. By writing that insertion sort's running time is $\Theta(n^2)$, we abstracted away some details of this function. Because asymptotic notation applies to functions, what we were writing as $\Theta(n^2)$ was the function $an^2 + bn + c$, which in that case happened to characterize the worst-case running time of insertion sort.

In this book, the functions to which we apply asymptotic notation will usually characterize the running times of algorithms. But asymptotic notation can apply to functions that characterize some other aspect of algorithms (the amount of space they use, for example), or even to functions that have nothing whatsoever to do with algorithms.

Even when we use asymptotic notation to apply to the running time of an algorithm, we need to understand *which* running time we mean. Sometimes we are interested in the worst-case running time. Often, however, we wish to characterize the running time no matter what the input. In other words, we often wish to make a blanket statement that covers all inputs, not just the worst case. We shall see asymptotic notations that are well suited to characterizing running times no matter what the input.

Θ -notation

In Chapter 2, we found that the worst-case running time of insertion sort is $T(n) = \Theta(n^2)$. Let us define what this notation means. For a given function g(n), we denote by $\Theta(g(n))$ the set of functions

 $\Theta(g(n)) = \{ f(n) : \text{ there exist positive constants } c_1, c_2, \text{ and } n_0 \text{ such that} \\ 0 \le c_1 g(n) \le f(n) \le c_2 g(n) \text{ for all } n \ge n_0 \} .^1$

¹Within set notation, a colon means "such that."



Figure 3.1 Graphic examples of the Θ , O, and Ω notations. In each part, the value of n_0 shown is the minimum possible value; any greater value would also work. (a) Θ -notation bounds a function to within constant factors. We write $f(n) = \Theta(g(n))$ if there exist positive constants n_0, c_1 , and c_2 such that at and to the right of n_0 , the value of f(n) always lies between $c_1g(n)$ and $c_2g(n)$ inclusive. (b) O-notation gives an upper bound for a function to within a constant factor. We write f(n) = O(g(n)) if there are positive constants n_0 and c such that at and to the right of n_0 , the value of f(n) always lies on or below cg(n). (c) Ω -notation gives a lower bound for a function to within a constant factor. We write a constant factor. We write $f(n) = \Omega(g(n))$ if there are positive constants n_0 and c such that at and to the right of n_0 , the value of f(n) always lies on or below cg(n). (c) Ω -notation gives a lower bound for a function to within a constant factor. We write $f(n) = \Omega(g(n))$ if there are positive constants n_0 and c such that at and to the right of n_0 , the value of f(n) always lies on or above cg(n).

A function f(n) belongs to the set $\Theta(g(n))$ if there exist positive constants c_1 and c_2 such that it can be "sandwiched" between $c_1g(n)$ and $c_2g(n)$, for sufficiently large n. Because $\Theta(g(n))$ is a set, we could write " $f(n) \in \Theta(g(n))$ " to indicate that f(n) is a member of $\Theta(g(n))$. Instead, we will usually write " $f(n) = \Theta(g(n))$ " to express the same notion. You might be confused because we abuse equality in this way, but we shall see later in this section that doing so has its advantages.

Figure 3.1(a) gives an intuitive picture of functions f(n) and g(n), where $f(n) = \Theta(g(n))$. For all values of n at and to the right of n_0 , the value of f(n) lies at or above $c_1g(n)$ and at or below $c_2g(n)$. In other words, for all $n \ge n_0$, the function f(n) is equal to g(n) to within a constant factor. We say that g(n) is an *asymptotically tight bound* for f(n).

The definition of $\Theta(g(n))$ requires that every member $f(n) \in \Theta(g(n))$ be *asymptotically nonnegative*, that is, that f(n) be nonnegative whenever n is sufficiently large. (An *asymptotically positive* function is one that is positive for all sufficiently large n.) Consequently, the function g(n) itself must be asymptotically nonnegative, or else the set $\Theta(g(n))$ is empty. We shall therefore assume that every function used within Θ -notation is asymptotically nonnegative. This assumption holds for the other asymptotic notations defined in this chapter as well.

In Chapter 2, we introduced an informal notion of Θ -notation that amounted to throwing away lower-order terms and ignoring the leading coefficient of the highest-order term. Let us briefly justify this intuition by using the formal definition to show that $\frac{1}{2}n^2 - 3n = \Theta(n^2)$. To do so, we must determine positive constants c_1 , c_2 , and n_0 such that

$$c_1 n^2 \le \frac{1}{2}n^2 - 3n \le c_2 n^2$$

for all $n \ge n_0$. Dividing by n^2 yields

$$c_1 \leq \frac{1}{2} - \frac{3}{n} \leq c_2$$
.

We can make the right-hand inequality hold for any value of $n \ge 1$ by choosing any constant $c_2 \ge 1/2$. Likewise, we can make the left-hand inequality hold for any value of $n \ge 7$ by choosing any constant $c_1 \le 1/14$. Thus, by choosing $c_1 = 1/14$, $c_2 = 1/2$, and $n_0 = 7$, we can verify that $\frac{1}{2}n^2 - 3n = \Theta(n^2)$. Certainly, other choices for the constants exist, but the important thing is that *some* choice exists. Note that these constants depend on the function $\frac{1}{2}n^2 - 3n$; a different function belonging to $\Theta(n^2)$ would usually require different constants.

We can also use the formal definition to verify that $6n^3 \neq \Theta(n^2)$. Suppose for the purpose of contradiction that c_2 and n_0 exist such that $6n^3 \leq c_2n^2$ for all $n \geq n_0$. But then dividing by n^2 yields $n \leq c_2/6$, which cannot possibly hold for arbitrarily large n, since c_2 is constant.

Intuitively, the lower-order terms of an asymptotically positive function can be ignored in determining asymptotically tight bounds because they are insignificant for large n. When n is large, even a tiny fraction of the highest-order term suffices to dominate the lower-order terms. Thus, setting c_1 to a value that is slightly smaller than the coefficient of the highest-order term and setting c_2 to a value that is slightly larger permits the inequalities in the definition of Θ -notation to be satisfied. The coefficient of the highest-order term can likewise be ignored, since it only changes c_1 and c_2 by a constant factor equal to the coefficient.

As an example, consider any quadratic function $f(n) = an^2 + bn + c$, where a, b, and c are constants and a > 0. Throwing away the lower-order terms and ignoring the constant yields $f(n) = \Theta(n^2)$. Formally, to show the same thing, we take the constants $c_1 = a/4$, $c_2 = 7a/4$, and $n_0 = 2 \cdot \max(|b|/a, \sqrt{|c|/a})$. You may verify that $0 \le c_1n^2 \le an^2 + bn + c \le c_2n^2$ for all $n \ge n_0$. In general, for any polynomial $p(n) = \sum_{i=0}^{d} a_i n^i$, where the a_i are constants and $a_d > 0$, we have $p(n) = \Theta(n^d)$ (see Problem 3-1).

Since any constant is a degree-0 polynomial, we can express any constant function as $\Theta(n^0)$, or $\Theta(1)$. This latter notation is a minor abuse, however, because the
expression does not indicate what variable is tending to infinity.² We shall often use the notation $\Theta(1)$ to mean either a constant or a constant function with respect to some variable.

O-notation

The Θ -notation asymptotically bounds a function from above and below. When we have only an *asymptotic upper bound*, we use *O*-notation. For a given function g(n), we denote by O(g(n)) (pronounced "big-oh of g of n" or sometimes just "oh of g of n") the set of functions

 $O(g(n)) = \{ f(n) : \text{ there exist positive constants } c \text{ and } n_0 \text{ such that} \\ 0 \le f(n) \le cg(n) \text{ for all } n \ge n_0 \}.$

We use *O*-notation to give an upper bound on a function, to within a constant factor. Figure 3.1(b) shows the intuition behind *O*-notation. For all values *n* at and to the right of n_0 , the value of the function f(n) is on or below cg(n).

We write f(n) = O(g(n)) to indicate that a function f(n) is a member of the set O(g(n)). Note that $f(n) = \Theta(g(n))$ implies f(n) = O(g(n)), since Θ -notation is a stronger notion than O-notation. Written set-theoretically, we have $\Theta(g(n)) \subseteq O(g(n))$. Thus, our proof that any quadratic function $an^2 + bn + c$, where a > 0, is in $\Theta(n^2)$ also shows that any such quadratic function is in $O(n^2)$. What may be more surprising is that when a > 0, any *linear* function an + b is in $O(n^2)$, which is easily verified by taking c = a + |b| and $n_0 = \max(1, -b/a)$.

If you have seen O-notation before, you might find it strange that we should write, for example, $n = O(n^2)$. In the literature, we sometimes find O-notation informally describing asymptotically tight bounds, that is, what we have defined using Θ -notation. In this book, however, when we write f(n) = O(g(n)), we are merely claiming that some constant multiple of g(n) is an asymptotic upper bound on f(n), with no claim about how tight an upper bound it is. Distinguishing asymptotic upper bounds from asymptotically tight bounds is standard in the algorithms literature.

Using *O*-notation, we can often describe the running time of an algorithm merely by inspecting the algorithm's overall structure. For example, the doubly nested loop structure of the insertion sort algorithm from Chapter 2 immediately yields an $O(n^2)$ upper bound on the worst-case running time: the cost of each iteration of the inner loop is bounded from above by O(1) (constant), the indices *i*

²The real problem is that our ordinary notation for functions does not distinguish functions from values. In λ -calculus, the parameters to a function are clearly specified: the function n^2 could be written as $\lambda n.n^2$, or even $\lambda r.r^2$. Adopting a more rigorous notation, however, would complicate algebraic manipulations, and so we choose to tolerate the abuse.



and j are both at most n, and the inner loop is executed at most once for each of the n^2 pairs of values for i and j.

Since *O*-notation describes an upper bound, when we use it to bound the worstcase running time of an algorithm, we have a bound on the running time of the algorithm on every input—the blanket statement we discussed earlier. Thus, the $O(n^2)$ bound on worst-case running time of insertion sort also applies to its running time on every input. The $\Theta(n^2)$ bound on the worst-case running time of insertion sort, however, does not imply a $\Theta(n^2)$ bound on the running time of insertion sort on *every* input. For example, we saw in Chapter 2 that when the input is already sorted, insertion sort runs in $\Theta(n)$ time.

Technically, it is an abuse to say that the running time of insertion sort is $O(n^2)$, since for a given *n*, the actual running time varies, depending on the particular input of size *n*. When we say "the running time is $O(n^2)$," we mean that there is a function f(n) that is $O(n^2)$ such that for any value of *n*, no matter what particular input of size *n* is chosen, the running time on that input is bounded from above by the value f(n). Equivalently, we mean that the worst-case running time is $O(n^2)$.

Ω -notation

Just as *O*-notation provides an asymptotic *upper* bound on a function, Ω -notation provides an *asymptotic lower bound*. For a given function g(n), we denote by $\Omega(g(n))$ (pronounced "big-omega of g of n" or sometimes just "omega of g of n") the set of functions

 $\Omega(g(n)) = \{ f(n) : \text{ there exist positive constants } c \text{ and } n_0 \text{ such that} \\ 0 \le cg(n) \le f(n) \text{ for all } n \ge n_0 \}.$

Figure 3.1(c) shows the intuition behind Ω -notation. For all values *n* at or to the right of n_0 , the value of f(n) is on or above cg(n).

From the definitions of the asymptotic notations we have seen thus far, it is easy to prove the following important theorem (see Exercise 3.1-5).

Theorem 3.1

For any two functions f(n) and g(n), we have $f(n) = \Theta(g(n))$ if and only if f(n) = O(g(n)) and $f(n) = \Omega(g(n))$.

As an example of the application of this theorem, our proof that $an^2 + bn + c = \Theta(n^2)$ for any constants a, b, and c, where a > 0, immediately implies that $an^2 + bn + c = \Omega(n^2)$ and $an^2 + bn + c = O(n^2)$. In practice, rather than using Theorem 3.1 to obtain asymptotic upper and lower bounds from asymptotically tight bounds, as we did for this example, we usually use it to prove asymptotically tight bounds from asymptotic upper and lower bounds.

When we say that the *running time* (no modifier) of an algorithm is $\Omega(g(n))$, we mean that *no matter what particular input of size n is chosen for each value of n*, the running time on that input is at least a constant times g(n), for sufficiently large *n*. Equivalently, we are giving a lower bound on the best-case running time of an algorithm. For example, the best-case running time of insertion sort is $\Omega(n)$, which implies that the running time of insertion sort is $\Omega(n)$.

The running time of insertion sort therefore belongs to both $\Omega(n)$ and $O(n^2)$, since it falls anywhere between a linear function of n and a quadratic function of n. Moreover, these bounds are asymptotically as tight as possible: for instance, the running time of insertion sort is not $\Omega(n^2)$, since there exists an input for which insertion sort runs in $\Theta(n)$ time (e.g., when the input is already sorted). It is not contradictory, however, to say that the *worst-case* running time of insertion sort is $\Omega(n^2)$, since there exists an input that causes the algorithm to take $\Omega(n^2)$ time.

Asymptotic notation in equations and inequalities

We have already seen how asymptotic notation can be used within mathematical formulas. For example, in introducing *O*-notation, we wrote " $n = O(n^2)$." We might also write $2n^2 + 3n + 1 = 2n^2 + \Theta(n)$. How do we interpret such formulas?

When the asymptotic notation stands alone (that is, not within a larger formula) on the right-hand side of an equation (or inequality), as in $n = O(n^2)$, we have already defined the equal sign to mean set membership: $n \in O(n^2)$. In general, however, when asymptotic notation appears in a formula, we interpret it as standing for some anonymous function that we do not care to name. For example, the formula $2n^2 + 3n + 1 = 2n^2 + \Theta(n)$ means that $2n^2 + 3n + 1 = 2n^2 + f(n)$, where f(n) is some function in the set $\Theta(n)$. In this case, we let f(n) = 3n + 1, which indeed is in $\Theta(n)$.

Using asymptotic notation in this manner can help eliminate inessential detail and clutter in an equation. For example, in Chapter 2 we expressed the worst-case running time of merge sort as the recurrence

$$T(n) = 2T(n/2) + \Theta(n) .$$

If we are interested only in the asymptotic behavior of T(n), there is no point in specifying all the lower-order terms exactly; they are all understood to be included in the anonymous function denoted by the term $\Theta(n)$.

The number of anonymous functions in an expression is understood to be equal to the number of times the asymptotic notation appears. For example, in the expression

$$\sum_{i=1}^n O(i)$$

there is only a single anonymous function (a function of *i*). This expression is thus *not* the same as $O(1) + O(2) + \cdots + O(n)$, which doesn't really have a clean interpretation.

In some cases, asymptotic notation appears on the left-hand side of an equation, as in

$$2n^2 + \Theta(n) = \Theta(n^2) \, .$$

We interpret such equations using the following rule: No matter how the anonymous functions are chosen on the left of the equal sign, there is a way to choose the anonymous functions on the right of the equal sign to make the equation valid. Thus, our example means that for any function $f(n) \in \Theta(n)$, there is some function $g(n) \in \Theta(n^2)$ such that $2n^2 + f(n) = g(n)$ for all n. In other words, the right-hand side of an equation provides a coarser level of detail than the left-hand side.

We can chain together a number of such relationships, as in

$$2n^2 + 3n + 1 = 2n^2 + \Theta(n)$$

= $\Theta(n^2)$.

We can interpret each equation separately by the rules above. The first equation says that there is *some* function $f(n) \in \Theta(n)$ such that $2n^2 + 3n + 1 = 2n^2 + f(n)$ for all n. The second equation says that for any function $g(n) \in \Theta(n)$ (such as the f(n) just mentioned), there is *some* function $h(n) \in \Theta(n^2)$ such that $2n^2 + g(n) = h(n)$ for all n. Note that this interpretation implies that $2n^2 + 3n + 1 = \Theta(n^2)$, which is what the chaining of equations intuitively gives us.

o-notation

The asymptotic upper bound provided by *O*-notation may or may not be asymptotically tight. The bound $2n^2 = O(n^2)$ is asymptotically tight, but the bound $2n = O(n^2)$ is not. We use *o*-notation to denote an upper bound that is not asymptotically tight. We formally define o(g(n)) ("little-oh of g of n") as the set

 $o(g(n)) = \{ f(n) : \text{ for any positive constant } c > 0, \text{ there exists a constant} \\ n_0 > 0 \text{ such that } 0 \le f(n) < cg(n) \text{ for all } n \ge n_0 \}.$

For example, $2n = o(n^2)$, but $2n^2 \neq o(n^2)$.

The definitions of *O*-notation and *o*-notation are similar. The main difference is that in f(n) = O(g(n)), the bound $0 \le f(n) \le cg(n)$ holds for *some* constant c > 0, but in f(n) = o(g(n)), the bound $0 \le f(n) < cg(n)$ holds for *all* constants c > 0. Intuitively, in *o*-notation, the function f(n) becomes insignificant relative to g(n) as *n* approaches infinity; that is,

$$\lim_{n \to \infty} \frac{f(n)}{g(n)} = 0.$$
(3.1)

Some authors use this limit as a definition of the *o*-notation; the definition in this book also restricts the anonymous functions to be asymptotically nonnegative.

ω -notation

By analogy, ω -notation is to Ω -notation as o-notation is to O-notation. We use ω -notation to denote a lower bound that is not asymptotically tight. One way to define it is by

$$f(n) \in \omega(g(n))$$
 if and only if $g(n) \in o(f(n))$.

Formally, however, we define $\omega(g(n))$ ("little-omega of g of n") as the set

 $\omega(g(n)) = \{f(n) : \text{ for any positive constant } c > 0, \text{ there exists a constant} \\ n_0 > 0 \text{ such that } 0 \le cg(n) < f(n) \text{ for all } n \ge n_0 \}.$

For example, $n^2/2 = \omega(n)$, but $n^2/2 \neq \omega(n^2)$. The relation $f(n) = \omega(g(n))$ implies that

$$\lim_{n \to \infty} \frac{f(n)}{g(n)} = \infty$$

if the limit exists. That is, f(n) becomes arbitrarily large relative to g(n) as n approaches infinity.

Comparing functions

Many of the relational properties of real numbers apply to asymptotic comparisons as well. For the following, assume that f(n) and g(n) are asymptotically positive.

Transitivity:

$$\begin{split} f(n) &= \Theta(g(n)) \text{ and } g(n) = \Theta(h(n)) & \text{imply} \quad f(n) = \Theta(h(n)) ,\\ f(n) &= O(g(n)) \text{ and } g(n) = O(h(n)) & \text{imply} \quad f(n) = O(h(n)) ,\\ f(n) &= \Omega(g(n)) \text{ and } g(n) = \Omega(h(n)) & \text{imply} \quad f(n) = \Omega(h(n)) ,\\ f(n) &= o(g(n)) \text{ and } g(n) = o(h(n)) & \text{imply} \quad f(n) = o(h(n)) ,\\ f(n) &= \omega(g(n)) \text{ and } g(n) = \omega(h(n)) & \text{imply} \quad f(n) = \omega(h(n)) . \end{split}$$

Reflexivity:

$$f(n) = \Theta(f(n)),$$

$$f(n) = O(f(n)),$$

$$f(n) = \Omega(f(n)).$$

Symmetry:

$$f(n) = \Theta(g(n))$$
 if and only if $g(n) = \Theta(f(n))$.

Transpose symmetry:

$$f(n) = O(g(n))$$
 if and only if $g(n) = \Omega(f(n))$,
 $f(n) = o(g(n))$ if and only if $g(n) = \omega(f(n))$.

Because these properties hold for asymptotic notations, we can draw an analogy between the asymptotic comparison of two functions f and g and the comparison of two real numbers a and b:

f(n) = O(g(n))	is like	$a\leq b$,
$f(n) = \Omega(g(n))$	is like	$a \ge b$,
$f(n) = \Theta(g(n))$	is like	a=b,
f(n) = o(g(n))	is like	a < b,
$f(n) = \omega(g(n))$	is like	a > b .

We say that f(n) is *asymptotically smaller* than g(n) if f(n) = o(g(n)), and f(n) is *asymptotically larger* than g(n) if $f(n) = \omega(g(n))$.

One property of real numbers, however, does not carry over to asymptotic notation:

Trichotomy: For any two real numbers *a* and *b*, exactly one of the following must hold: a < b, a = b, or a > b.

Although any two real numbers can be compared, not all functions are asymptotically comparable. That is, for two functions f(n) and g(n), it may be the case that neither f(n) = O(g(n)) nor $f(n) = \Omega(g(n))$ holds. For example, we cannot compare the functions n and $n^{1+\sin n}$ using asymptotic notation, since the value of the exponent in $n^{1+\sin n}$ oscillates between 0 and 2, taking on all values in between.

Exercises

3.1-1

Let f(n) and g(n) be asymptotically nonnegative functions. Using the basic definition of Θ -notation, prove that $\max(f(n), g(n)) = \Theta(f(n) + g(n))$.

(3.2)

3.1-2

Show that for any real constants a and b, where b > 0,

$$(n+a)^b = \Theta(n^b) \ .$$

3.1-3

Explain why the statement, "The running time of algorithm A is at least $O(n^2)$," is meaningless.

3.1-4 Is $2^{n+1} = O(2^n)$? Is $2^{2n} = O(2^n)$?

3.1-5

Prove Theorem 3.1.

3.1-6

Prove that the running time of an algorithm is $\Theta(g(n))$ if and only if its worst-case running time is O(g(n)) and its best-case running time is $\Omega(g(n))$.

3.1-7

Prove that $o(g(n)) \cap \omega(g(n))$ is the empty set.

3.1-8

We can extend our notation to the case of two parameters n and m that can go to infinity independently at different rates. For a given function g(n,m), we denote by O(g(n,m)) the set of functions

 $O(g(n,m)) = \{f(n,m) : \text{ there exist positive constants } c, n_0, \text{ and } m_0 \\ \text{ such that } 0 \le f(n,m) \le cg(n,m) \\ \text{ for all } n \ge n_0 \text{ or } m \ge m_0 \}.$

Give corresponding definitions for $\Omega(g(n,m))$ and $\Theta(g(n,m))$.

3.2 Standard notations and common functions

This section reviews some standard mathematical functions and notations and explores the relationships among them. It also illustrates the use of the asymptotic notations.

Monotonicity

A function f(n) is *monotonically increasing* if $m \le n$ implies $f(m) \le f(n)$. Similarly, it is *monotonically decreasing* if $m \le n$ implies $f(m) \ge f(n)$. A function f(n) is *strictly increasing* if m < n implies f(m) < f(n) and *strictly decreasing* if m < n implies f(m) < f(n) and *strictly decreasing* if m < n implies f(m) > f(n).

Floors and ceilings

For any real number *x*, we denote the greatest integer less than or equal to *x* by $\lfloor x \rfloor$ (read "the floor of *x*") and the least integer greater than or equal to *x* by $\lceil x \rceil$ (read "the ceiling of *x*"). For all real *x*,

$$x - 1 < \lfloor x \rfloor \le x \le \lceil x \rceil < x + 1.$$

$$(3.3)$$

For any integer n,

$$\lceil n/2 \rceil + \lfloor n/2 \rfloor = n ,$$

and for any real number $x \ge 0$ and integers a, b > 0,

$$\begin{bmatrix} \frac{\lceil x/a \rceil}{b} \end{bmatrix} = \begin{bmatrix} \frac{x}{ab} \end{bmatrix}, \tag{3.4}$$

$$\left\lfloor \frac{\lfloor x/a \rfloor}{b} \right\rfloor = \left\lfloor \frac{x}{ab} \right\rfloor, \tag{3.5}$$

$$\left\lceil \frac{a}{b} \right\rceil \leq \frac{a + (b - 1)}{b}, \qquad (3.6)$$

$$\left\lfloor \frac{a}{b} \right\rfloor \geq \frac{a - (b - 1)}{b} \,. \tag{3.7}$$

The floor function $f(x) = \lfloor x \rfloor$ is monotonically increasing, as is the ceiling function $f(x) = \lceil x \rceil$.

Modular arithmetic

For any integer *a* and any positive integer *n*, the value *a* mod *n* is the *remainder* (or *residue*) of the quotient a/n:

$$a \mod n = a - n \lfloor a/n \rfloor . \tag{3.8}$$

It follows that

$$0 \le a \mod n < n \ . \tag{3.9}$$

Given a well-defined notion of the remainder of one integer when divided by another, it is convenient to provide special notation to indicate equality of remainders. If $(a \mod n) = (b \mod n)$, we write $a \equiv b \pmod{n}$ and say that *a* is *equivalent* to *b*, modulo *n*. In other words, $a \equiv b \pmod{n}$ if *a* and *b* have the same remainder when divided by *n*. Equivalently, $a \equiv b \pmod{n}$ if and only if *n* is a divisor of b - a. We write $a \neq b \pmod{n}$ if *a* is not equivalent to *b*, modulo *n*.

Polynomials

Given a nonnegative integer d, a *polynomial in n of degree d* is a function p(n) of the form

$$p(n) = \sum_{i=0}^{d} a_i n^i$$

where the constants a_0, a_1, \ldots, a_d are the **coefficients** of the polynomial and $a_d \neq 0$. A polynomial is asymptotically positive if and only if $a_d > 0$. For an asymptotically positive polynomial p(n) of degree d, we have $p(n) = \Theta(n^d)$. For any real constant $a \ge 0$, the function n^a is monotonically increasing, and for any real constant $a \le 0$, the function n^a is monotonically decreasing. We say that a function f(n) is **polynomially bounded** if $f(n) = O(n^k)$ for some constant k.

Exponentials

0

For all real a > 0, m, and n, we have the following identities:

$$a^{0} = 1,$$

$$a^{1} = a,$$

$$a^{-1} = 1/a,$$

$$(a^{m})^{n} = a^{mn},$$

$$(a^{m})^{n} = (a^{n})^{m},$$

$$a^{m}a^{n} = a^{m+n}.$$

For all *n* and $a \ge 1$, the function a^n is monotonically increasing in *n*. When convenient, we shall assume $0^0 = 1$.

We can relate the rates of growth of polynomials and exponentials by the following fact. For all real constants a and b such that a > 1,

$$\lim_{n \to \infty} \frac{n^b}{a^n} = 0 , \qquad (3.10)$$

from which we can conclude that

$$n^b = o(a^n)$$

Thus, any exponential function with a base strictly greater than 1 grows faster than any polynomial function.

Using *e* to denote 2.71828..., the base of the natural logarithm function, we have for all real *x*,

$$e^{x} = 1 + x + \frac{x^{2}}{2!} + \frac{x^{3}}{3!} + \dots = \sum_{i=0}^{\infty} \frac{x^{i}}{i!},$$
 (3.11)

where "!" denotes the factorial function defined later in this section. For all real x, we have the inequality

$$e^x \ge 1 + x , \tag{3.12}$$

where equality holds only when x = 0. When $|x| \le 1$, we have the approximation

$$1 + x \le e^x \le 1 + x + x^2 \,. \tag{3.13}$$

When $x \to 0$, the approximation of e^x by 1 + x is quite good:

$$e^x = 1 + x + \Theta(x^2) \,.$$

(In this equation, the asymptotic notation is used to describe the limiting behavior as $x \to 0$ rather than as $x \to \infty$.) We have for all x,

$$\lim_{n \to \infty} \left(1 + \frac{x}{n} \right)^n = e^x .$$
(3.14)

Logarithms

We shall use the following notations:

 $lg n = log_2 n \quad (binary logarithm) ,$ $ln n = log_e n \quad (natural logarithm) ,$ $lg^k n = (lg n)^k \quad (exponentiation) ,$ $lg lg n = lg(lg n) \quad (composition) .$

An important notational convention we shall adopt is that *logarithm functions will* apply only to the next term in the formula, so that $\lg n + k$ will mean $(\lg n) + k$ and not $\lg(n + k)$. If we hold b > 1 constant, then for n > 0, the function $\log_b n$ is strictly increasing.

For all real a > 0, b > 0, c > 0, and n,

$$a = b^{\log_b a},$$

$$\log_c(ab) = \log_c a + \log_c b,$$

$$\log_b a^n = n \log_b a,$$

$$\log_b a = \frac{\log_c a}{\log_c b},$$

$$\log_b (1/a) = -\log_b a,$$

$$\log_b a = \frac{1}{\log_a b},$$

$$a^{\log_b c} = c^{\log_b a},$$

(3.15)

where, in each equation above, logarithm bases are not 1.

By equation (3.15), changing the base of a logarithm from one constant to another changes the value of the logarithm by only a constant factor, and so we shall often use the notation " $\lg n$ " when we don't care about constant factors, such as in *O*-notation. Computer scientists find 2 to be the most natural base for logarithms because so many algorithms and data structures involve splitting a problem into two parts.

There is a simple series expansion for $\ln(1 + x)$ when |x| < 1:

$$\ln(1+x) = x - \frac{x^2}{2} + \frac{x^3}{3} - \frac{x^4}{4} + \frac{x^5}{5} - \cdots$$

We also have the following inequalities for x > -1:

$$\frac{x}{1+x} \le \ln(1+x) \le x , \qquad (3.17)$$

where equality holds only for x = 0.

We say that a function f(n) is **polylogarithmically bounded** if $f(n) = O(\lg^k n)$ for some constant k. We can relate the growth of polynomials and polylogarithms by substituting $\lg n$ for n and 2^a for a in equation (3.10), yielding

$$\lim_{n \to \infty} \frac{\lg^b n}{(2^a)^{\lg n}} = \lim_{n \to \infty} \frac{\lg^b n}{n^a} = 0$$

From this limit, we can conclude that

$$\lg^b n = o(n^a)$$

for any constant a > 0. Thus, any positive polynomial function grows faster than any polylogarithmic function.

Factorials

The notation n! (read "*n* factorial") is defined for integers $n \ge 0$ as

$$n! = \begin{cases} 1 & \text{if } n = 0 \\ n \cdot (n-1)! & \text{if } n > 0 \end{cases}$$

Thus, $n! = 1 \cdot 2 \cdot 3 \cdots n$.

A weak upper bound on the factorial function is $n! \leq n^n$, since each of the *n* terms in the factorial product is at most *n*. *Stirling's approximation*,

$$n! = \sqrt{2\pi n} \left(\frac{n}{e}\right)^n \left(1 + \Theta\left(\frac{1}{n}\right)\right) , \qquad (3.18)$$

where e is the base of the natural logarithm, gives us a tighter upper bound, and a lower bound as well. As Exercise 3.2-3 asks you to prove,

$$n! = o(n^{n}),$$

$$n! = \omega(2^{n}),$$

$$\lg(n!) = \Theta(n \lg n),$$
(3.19)

where Stirling's approximation is helpful in proving equation (3.19). The following equation also holds for all $n \ge 1$:

$$n! = \sqrt{2\pi n} \left(\frac{n}{e}\right)^n e^{\alpha_n} \tag{3.20}$$

where

$$\frac{1}{12n+1} < \alpha_n < \frac{1}{12n} \,. \tag{3.21}$$

Functional iteration

We use the notation $f^{(i)}(n)$ to denote the function f(n) iteratively applied *i* times to an initial value of *n*. Formally, let f(n) be a function over the reals. For non-negative integers *i*, we recursively define

$$f^{(i)}(n) = \begin{cases} n & \text{if } i = 0, \\ f(f^{(i-1)}(n)) & \text{if } i > 0. \end{cases}$$

For example, if f(n) = 2n, then $f^{(i)}(n) = 2^i n$.

The iterated logarithm function

We use the notation $\lg^* n$ (read "log star of n") to denote the iterated logarithm, defined as follows. Let $\lg^{(i)} n$ be as defined above, with $f(n) = \lg n$. Because the logarithm of a nonpositive number is undefined, $\lg^{(i)} n$ is defined only if $\lg^{(i-1)} n > 0$. Be sure to distinguish $\lg^{(i)} n$ (the logarithm function applied *i* times in succession, starting with argument *n*) from $\lg^i n$ (the logarithm of *n* raised to the *i*th power). Then we define the iterated logarithm function as

 $\lg^* n = \min \{ i \ge 0 : \lg^{(i)} n \le 1 \} .$

The iterated logarithm is a very slowly growing function:

```
\begin{split} lg^* 2 &= 1 , \\ lg^* 4 &= 2 , \\ lg^* 16 &= 3 , \\ lg^* 65536 &= 4 , \\ lg^* (2^{65536}) &= 5 . \end{split}
```

Since the number of atoms in the observable universe is estimated to be about 10^{80} , which is much less than 2^{65536} , we rarely encounter an input size *n* such that $\lg^* n > 5$.

Fibonacci numbers

We define the *Fibonacci numbers* by the following recurrence:

$$F_{0} = 0,$$

$$F_{1} = 1,$$

$$F_{i} = F_{i-1} + F_{i-2} \quad \text{for } i \ge 2.$$
(3.22)

Thus, each Fibonacci number is the sum of the two previous ones, yielding the sequence

.

Fibonacci numbers are related to the *golden ratio* ϕ and to its conjugate $\hat{\phi}$, which are the two roots of the equation

$$x^2 = x + 1 \tag{3.23}$$

and are given by the following formulas (see Exercise 3.2-6):

$$\phi = \frac{1 + \sqrt{5}}{2}$$

$$= 1.61803...,$$

$$\hat{\phi} = \frac{1 - \sqrt{5}}{2}$$

$$= -.61803....$$
(3.24)

Specifically, we have

$$F_i = \frac{\phi^i - \hat{\phi}^i}{\sqrt{5}} \,,$$

which we can prove by induction (Exercise 3.2-7). Since $|\hat{\phi}| < 1$, we have

$$\begin{array}{c|c} |\widehat{\phi}^i| \\ \hline \sqrt{5} & < & \frac{1}{\sqrt{5}} \\ & < & \frac{1}{2} \end{array} ,$$

which implies that

$$F_i = \left\lfloor \frac{\phi^i}{\sqrt{5}} + \frac{1}{2} \right\rfloor, \tag{3.25}$$

which is to say that the *i*th Fibonacci number F_i is equal to $\phi^i / \sqrt{5}$ rounded to the nearest integer. Thus, Fibonacci numbers grow exponentially.

Exercises

3.2-1

Show that if f(n) and g(n) are monotonically increasing functions, then so are the functions f(n) + g(n) and f(g(n)), and if f(n) and g(n) are in addition nonnegative, then $f(n) \cdot g(n)$ is monotonically increasing.

3.2-2

Prove equation (3.16).

3.2-3

Prove equation (3.19). Also prove that $n! = \omega(2^n)$ and $n! = o(n^n)$.

3.2-4 **★**

Is the function $\lceil \lg n \rceil!$ polynomially bounded? Is the function $\lceil \lg \lg n \rceil!$ polynomially bounded?

3.2-5 *

Which is asymptotically larger: $lg(lg^*n)$ or $lg^*(lgn)$?

3.2-6

Show that the golden ratio ϕ and its conjugate $\hat{\phi}$ both satisfy the equation $x^2 = x + 1$.

3.2-7

Prove by induction that the *i*th Fibonacci number satisfies the equality

$$F_i = \frac{\phi^i - \widehat{\phi}^i}{\sqrt{5}} \,,$$

where ϕ is the golden ratio and $\hat{\phi}$ is its conjugate.

3.2-8

Show that $k \ln k = \Theta(n)$ implies $k = \Theta(n / \ln n)$.

Problems

3-1 Asymptotic behavior of polynomials Let

$$p(n) = \sum_{i=0}^{u} a_i n^i \; ,$$

where $a_d > 0$, be a degree-*d* polynomial in *n*, and let *k* be a constant. Use the definitions of the asymptotic notations to prove the following properties.

a. If
$$k \ge d$$
, then $p(n) = O(n^k)$.

- **b.** If $k \leq d$, then $p(n) = \Omega(n^k)$.
- c. If k = d, then $p(n) = \Theta(n^k)$.
- *d.* If k > d, then $p(n) = o(n^k)$.
- e. If k < d, then $p(n) = \omega(n^k)$.

3-2 Relative asymptotic growths

Indicate, for each pair of expressions (A, B) in the table below, whether A is O, o, Ω, ω , or Θ of B. Assume that $k \ge 1, \epsilon > 0$, and c > 1 are constants. Your answer should be in the form of the table with "yes" or "no" written in each box.

	A	В	0	0	Ω	ω	Θ
a.	$\lg^k n$	n^{ϵ}					
b.	n^k	c^n					
c.	\sqrt{n}	$n^{\sin n}$					
d.	2 ⁿ	$2^{n/2}$					
e.	$n^{\lg c}$	$C^{\lg n}$					
f.	lg(<i>n</i> !)	$lg(n^n)$					

3-3 Ordering by asymptotic growth rates

a. Rank the following functions by order of growth; that is, find an arrangement g_1, g_2, \ldots, g_{30} of the functions satisfying $g_1 = \Omega(g_2), g_2 = \Omega(g_3), \ldots, g_{29} = \Omega(g_{30})$. Partition your list into equivalence classes such that functions f(n) and g(n) are in the same class if and only if $f(n) = \Theta(g(n))$.

$\lg(\lg^* n)$	$2^{\lg^* n}$	$(\sqrt{2})^{\lg n}$	n^2	n!	(lg <i>n</i>)!
$(\frac{3}{2})^n$	n^3	$\lg^2 n$	lg(<i>n</i> !)	$2^{2^{n}}$	$n^{1/\lg n}$
ln ln <i>n</i>	$\lg^* n$	$n \cdot 2^n$	$n^{\lg \lg n}$	ln <i>n</i>	1
$2^{\lg n}$	$(\lg n)^{\lg n}$	e^n	$4^{\lg n}$	(n + 1)!	$\sqrt{\lg n}$
$\lg^*(\lg n)$	$2^{\sqrt{2 \lg n}}$	п	2^n	<i>n</i> lg <i>n</i>	$2^{2^{n+1}}$

b. Give an example of a single nonnegative function f(n) such that for all functions $g_i(n)$ in part (a), f(n) is neither $O(g_i(n))$ nor $\Omega(g_i(n))$.

3-4 Asymptotic notation properties

Let f(n) and g(n) be asymptotically positive functions. Prove or disprove each of the following conjectures.

- a. f(n) = O(g(n)) implies g(n) = O(f(n)).
- **b.** $f(n) + g(n) = \Theta(\min(f(n), g(n))).$
- c. f(n) = O(g(n)) implies $\lg(f(n)) = O(\lg(g(n)))$, where $\lg(g(n)) \ge 1$ and $f(n) \ge 1$ for all sufficiently large n.
- d. f(n) = O(g(n)) implies $2^{f(n)} = O(2^{g(n)})$.
- e. $f(n) = O((f(n))^2)$.
- f. f(n) = O(g(n)) implies $g(n) = \Omega(f(n))$.
- **g.** $f(n) = \Theta(f(n/2)).$
- **h.** $f(n) + o(f(n)) = \Theta(f(n))$.

3-5 Variations on O and Ω

Some authors define Ω in a slightly different way than we do; let's use $\widetilde{\Omega}$ (read "omega infinity") for this alternative definition. We say that $f(n) = \widetilde{\Omega}(g(n))$ if there exists a positive constant c such that $f(n) \ge cg(n) \ge 0$ for infinitely many integers n.

a. Show that for any two functions f(n) and g(n) that are asymptotically nonnegative, either f(n) = O(g(n)) or $f(n) = \tilde{\Omega}(g(n))$ or both, whereas this is not true if we use Ω in place of $\tilde{\Omega}$.

b. Describe the potential advantages and disadvantages of using $\tilde{\Omega}$ instead of Ω to characterize the running times of programs.

Some authors also define O in a slightly different manner; let's use O' for the alternative definition. We say that f(n) = O'(g(n)) if and only if |f(n)| = O(g(n)).

c. What happens to each direction of the "if and only if" in Theorem 3.1 if we substitute O' for O but still use Ω ?

Some authors define \widetilde{O} (read "soft-oh") to mean O with logarithmic factors ignored:

$$\widetilde{O}(g(n)) = \{f(n) : \text{ there exist positive constants } c, k, \text{ and } n_0 \text{ such that} \\ 0 \le f(n) \le cg(n) \lg^k(n) \text{ for all } n \ge n_0 \}.$$

d. Define $\widetilde{\Omega}$ and $\widetilde{\Theta}$ in a similar manner. Prove the corresponding analog to Theorem 3.1.

3-6 *Iterated functions*

We can apply the iteration operator * used in the lg* function to any monotonically increasing function f(n) over the reals. For a given constant $c \in \mathbb{R}$, we define the iterated function f_c^* by

$$f_c^*(n) = \min\{i \ge 0 : f^{(i)}(n) \le c\}$$
,

which need not be well defined in all cases. In other words, the quantity $f_c^*(n)$ is the number of iterated applications of the function f required to reduce its argument down to c or less.

For each of the following functions f(n) and constants c, give as tight a bound as possible on $f_c^*(n)$.

	f(n)	С	$f_c^*(n)$
a.	n - 1	0	
b.	lg <i>n</i>	1	
c.	n/2	1	
d.	<i>n</i> /2	2	
e.	\sqrt{n}	2	
f.	\sqrt{n}	1	
g.	$n^{1/3}$	2	
h.	$n/\lg n$	2	

Chapter notes

Knuth [209] traces the origin of the *O*-notation to a number-theory text by P. Bachmann in 1892. The *o*-notation was invented by E. Landau in 1909 for his discussion of the distribution of prime numbers. The Ω and Θ notations were advocated by Knuth [213] to correct the popular, but technically sloppy, practice in the literature of using *O*-notation for both upper and lower bounds. Many people continue to use the *O*-notation where the Θ -notation is more technically precise. Further discussion of the history and development of asymptotic notations appears in works by Knuth [209, 213] and Brassard and Bratley [54].

Not all authors define the asymptotic notations in the same way, although the various definitions agree in most common situations. Some of the alternative definitions encompass functions that are not asymptotically nonnegative, as long as their absolute values are appropriately bounded.

Equation (3.20) is due to Robbins [297]. Other properties of elementary mathematical functions can be found in any good mathematical reference, such as Abramowitz and Stegun [1] or Zwillinger [362], or in a calculus book, such as Apostol [18] or Thomas et al. [334]. Knuth [209] and Graham, Knuth, and Patashnik [152] contain a wealth of material on discrete mathematics as used in computer science.

4 Divide-and-Conquer

In Section 2.3.1, we saw how merge sort serves as an example of the divide-andconquer paradigm. Recall that in divide-and-conquer, we solve a problem recursively, applying three steps at each level of the recursion:

- **Divide** the problem into a number of subproblems that are smaller instances of the same problem.
- **Conquer** the subproblems by solving them recursively. If the subproblem sizes are small enough, however, just solve the subproblems in a straightforward manner.
- **Combine** the solutions to the subproblems into the solution for the original problem.

When the subproblems are large enough to solve recursively, we call that the *recursive case*. Once the subproblems become small enough that we no longer recurse, we say that the recursion "bottoms out" and that we have gotten down to the *base case*. Sometimes, in addition to subproblems that are smaller instances of the same problem, we have to solve subproblems that are not quite the same as the original problem. We consider solving such subproblems as part of the combine step.

In this chapter, we shall see more algorithms based on divide-and-conquer. The first one solves the maximum-subarray problem: it takes as input an array of numbers, and it determines the contiguous subarray whose values have the greatest sum. Then we shall see two divide-and-conquer algorithms for multiplying $n \times n$ matrices. One runs in $\Theta(n^3)$ time, which is no better than the straightforward method of multiplying square matrices. But the other, Strassen's algorithm, runs in $O(n^{2.81})$ time, which beats the straightforward method asymptotically.

Recurrences

Recurrences go hand in hand with the divide-and-conquer paradigm, because they give us a natural way to characterize the running times of divide-and-conquer algorithms. A *recurrence* is an equation or inequality that describes a function in terms

of its value on smaller inputs. For example, in Section 2.3.2 we described the worst-case running time T(n) of the MERGE-SORT procedure by the recurrence

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ 2T(n/2) + \Theta(n) & \text{if } n > 1 , \end{cases}$$
(4.1)

whose solution we claimed to be $T(n) = \Theta(n \lg n)$.

Recurrences can take many forms. For example, a recursive algorithm might divide subproblems into unequal sizes, such as a 2/3-to-1/3 split. If the divide and combine steps take linear time, such an algorithm would give rise to the recurrence $T(n) = T(2n/3) + T(n/3) + \Theta(n)$.

Subproblems are not necessarily constrained to being a constant fraction of the original problem size. For example, a recursive version of linear search (see Exercise 2.1-3) would create just one subproblem containing only one element fewer than the original problem. Each recursive call would take constant time plus the time for the recursive calls it makes, yielding the recurrence $T(n) = T(n-1) + \Theta(1)$.

This chapter offers three methods for solving recurrences—that is, for obtaining asymptotic " Θ " or "O" bounds on the solution:

- In the *substitution method*, we guess a bound and then use mathematical induction to prove our guess correct.
- The *recursion-tree method* converts the recurrence into a tree whose nodes represent the costs incurred at various levels of the recursion. We use techniques for bounding summations to solve the recurrence.
- The master method provides bounds for recurrences of the form

$$T(n) = aT(n/b) + f(n),$$
 (4.2)

where $a \ge 1$, b > 1, and f(n) is a given function. Such recurrences arise frequently. A recurrence of the form in equation (4.2) characterizes a divideand-conquer algorithm that creates a subproblems, each of which is 1/b the size of the original problem, and in which the divide and combine steps together take f(n) time.

To use the master method, you will need to memorize three cases, but once you do that, you will easily be able to determine asymptotic bounds for many simple recurrences. We will use the master method to determine the running times of the divide-and-conquer algorithms for the maximum-subarray problem and for matrix multiplication, as well as for other algorithms based on divideand-conquer elsewhere in this book.

Occasionally, we shall see recurrences that are not equalities but rather inequalities, such as $T(n) \leq 2T(n/2) + \Theta(n)$. Because such a recurrence states only an upper bound on T(n), we will couch its solution using *O*-notation rather than Θ -notation. Similarly, if the inequality were reversed to $T(n) \geq 2T(n/2) + \Theta(n)$, then because the recurrence gives only a lower bound on T(n), we would use Ω -notation in its solution.

Technicalities in recurrences

In practice, we neglect certain technical details when we state and solve recurrences. For example, if we call MERGE-SORT on *n* elements when *n* is odd, we end up with subproblems of size $\lfloor n/2 \rfloor$ and $\lceil n/2 \rceil$. Neither size is actually n/2, because n/2 is not an integer when *n* is odd. Technically, the recurrence describing the worst-case running time of MERGE-SORT is really

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ T(\lceil n/2 \rceil) + T(\lfloor n/2 \rfloor) + \Theta(n) & \text{if } n > 1 . \end{cases}$$
(4.3)

Boundary conditions represent another class of details that we typically ignore. Since the running time of an algorithm on a constant-sized input is a constant, the recurrences that arise from the running times of algorithms generally have $T(n) = \Theta(1)$ for sufficiently small n. Consequently, for convenience, we shall generally omit statements of the boundary conditions of recurrences and assume that T(n) is constant for small n. For example, we normally state recurrence (4.1) as

$$T(n) = 2T(n/2) + \Theta(n), \qquad (4.4)$$

without explicitly giving values for small n. The reason is that although changing the value of T(1) changes the exact solution to the recurrence, the solution typically doesn't change by more than a constant factor, and so the order of growth is unchanged.

When we state and solve recurrences, we often omit floors, ceilings, and boundary conditions. We forge ahead without these details and later determine whether or not they matter. They usually do not, but you should know when they do. Experience helps, and so do some theorems stating that these details do not affect the asymptotic bounds of many recurrences characterizing divide-and-conquer algorithms (see Theorem 4.1). In this chapter, however, we shall address some of these details and illustrate the fine points of recurrence solution methods.

4.1 The maximum-subarray problem

Suppose that you been offered the opportunity to invest in the Volatile Chemical Corporation. Like the chemicals the company produces, the stock price of the Volatile Chemical Corporation is rather volatile. You are allowed to buy one unit of stock only one time and then sell it at a later date, buying and selling after the close of trading for the day. To compensate for this restriction, you are allowed to learn what the price of the stock will be in the future. Your goal is to maximize your profit. Figure 4.1 shows the price of the stock over a 17-day period. You may buy the stock at any one time, starting after day 0, when the price is \$100 per share. Of course, you would want to "buy low, sell high"—buy at the lowest possible price and later on sell at the highest possible price and then sell at the highest price 4.1, the lowest price occurs after day 7, which occurs after the highest price, after day 1.

You might think that you can always maximize profit by either buying at the lowest price or selling at the highest price. For example, in Figure 4.1, we would maximize profit by buying at the lowest price, after day 7. If this strategy always worked, then it would be easy to determine how to maximize profit: find the highest and lowest prices, and then work left from the highest price to find the lowest price, work right from the lowest price to find the highest later price, and take the pair with the greater difference. Figure 4.2 shows a simple counterexample,



Figure 4.1 Information about the price of stock in the Volatile Chemical Corporation after the close of trading over a period of 17 days. The horizontal axis of the chart indicates the day, and the vertical axis shows the price. The bottom row of the table gives the change in price from the previous day.

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Figure 4.2 An example showing that the maximum profit does not always start at the lowest price or end at the highest price. Again, the horizontal axis indicates the day, and the vertical axis shows the price. Here, the maximum profit of \$3 per share would be earned by buying after day 2 and selling after day 3. The price of \$7 after day 2 is not the lowest price overall, and the price of \$10 after day 3 is not the highest price overall.

demonstrating that the maximum profit sometimes comes neither by buying at the lowest price nor by selling at the highest price.

A brute-force solution

We can easily devise a brute-force solution to this problem: just try every possible pair of buy and sell dates in which the buy date precedes the sell date. A period of n days has $\binom{n}{2}$ such pairs of dates. Since $\binom{n}{2}$ is $\Theta(n^2)$, and the best we can hope for is to evaluate each pair of dates in constant time, this approach would take $\Omega(n^2)$ time. Can we do better?

A transformation

In order to design an algorithm with an $o(n^2)$ running time, we will look at the input in a slightly different way. We want to find a sequence of days over which the net change from the first day to the last is maximum. Instead of looking at the daily prices, let us instead consider the daily change in price, where the change on day *i* is the difference between the prices after day i - 1 and after day *i*. The table in Figure 4.1 shows these daily changes in the bottom row. If we treat this row as an array *A*, shown in Figure 4.3, we now want to find the nonempty, contiguous subarray of *A* whose values have the largest sum. We call this contiguous subarray the *maximum subarray*. For example, in the array of Figure 4.3, the maximum subarray of A[1..16] is A[8..11], with the sum 43. Thus, you would want to buy the stock just before day 8 (that is, after day 7) and sell it after day 11, earning a profit of \$43 per share.

At first glance, this transformation does not help. We still need to check $\binom{n-1}{2} = \Theta(n^2)$ subarrays for a period of *n* days. Exercise 4.1-2 asks you to show



Figure 4.3 The change in stock prices as a maximum-subarray problem. Here, the subarray A[8..11], with sum 43, has the greatest sum of any contiguous subarray of array A.

that although computing the cost of one subarray might take time proportional to the length of the subarray, when computing all $\Theta(n^2)$ subarray sums, we can organize the computation so that each subarray sum takes O(1) time, given the values of previously computed subarray sums, so that the brute-force solution takes $\Theta(n^2)$ time.

So let us seek a more efficient solution to the maximum-subarray problem. When doing so, we will usually speak of "a" maximum subarray rather than "the" maximum subarray, since there could be more than one subarray that achieves the maximum sum.

The maximum-subarray problem is interesting only when the array contains some negative numbers. If all the array entries were nonnegative, then the maximum-subarray problem would present no challenge, since the entire array would give the greatest sum.

A solution using divide-and-conquer

Let's think about how we might solve the maximum-subarray problem using the divide-and-conquer technique. Suppose we want to find a maximum subarray of the subarray A[low..high]. Divide-and-conquer suggests that we divide the subarray into two subarrays of as equal size as possible. That is, we find the midpoint, say *mid*, of the subarray, and consider the subarrays A[low..mid]and A[mid + 1..high]. As Figure 4.4(a) shows, any contiguous subarray A[i..j]of A[low..high] must lie in exactly one of the following places:

- entirely in the subarray $A[low \dots mid]$, so that $low \le i \le j \le mid$,
- entirely in the subarray $A[mid + 1 \dots high]$, so that $mid < i \le j \le high$, or
- crossing the midpoint, so that $low \le i \le mid < j \le high$.

Therefore, a maximum subarray of A[low .. high] must lie in exactly one of these places. In fact, a maximum subarray of A[low .. high] must have the greatest sum over all subarrays entirely in A[low .. mid], entirely in A[mid + 1 .. high], or crossing the midpoint. We can find maximum subarrays of A[low .. mid] and A[mid+1.. high] recursively, because these two subproblems are smaller instances of the problem of finding a maximum subarray. Thus, all that is left to do is find a



Figure 4.4 (a) Possible locations of subarrays of A[low .. high]: entirely in A[low .. mid], entirely in A[mid + 1.. high], or crossing the midpoint *mid*. (b) Any subarray of A[low .. high] crossing the midpoint comprises two subarrays A[i .. mid] and A[mid + 1.. j], where $low \le i \le mid$ and $mid < j \le high$.

maximum subarray that crosses the midpoint, and take a subarray with the largest sum of the three.

We can easily find a maximum subarray crossing the midpoint in time linear in the size of the subarray A[low..high]. This problem is *not* a smaller instance of our original problem, because it has the added restriction that the subarray it chooses must cross the midpoint. As Figure 4.4(b) shows, any subarray crossing the midpoint is itself made of two subarrays A[i..mid] and A[mid + 1..j], where $low \le i \le mid$ and $mid < j \le high$. Therefore, we just need to find maximum subarrays of the form A[i..mid] and A[mid + 1..j] and then combine them. The procedure FIND-MAX-CROSSING-SUBARRAY takes as input the array A and the indices *low*, *mid*, and *high*, and it returns a tuple containing the indices demarcating a maximum subarray that crosses the midpoint, along with the sum of the values in a maximum subarray.

FIND-MAX-CROSSING-SUBARRAY (A, low, mid, high)

1 left-sum = $-\infty$ 2 sum = 03 for i = mid downto low 4 sum = sum + A[i]5 if sum > left-sum 6 left-sum = sum7 max-left = i8 right-sum = $-\infty$ sum = 09 10 for j = mid + 1 to high 11 sum = sum + A[j]12 **if** sum > right-sum 13 right-sum = sum 14 max-right = i15 **return** (*max-left*, *max-right*, *left-sum* + *right-sum*)

This procedure works as follows. Lines 1–7 find a maximum subarray of the left half, A[low..mid]. Since this subarray must contain A[mid], the **for** loop of lines 3–7 starts the index *i* at mid and works down to low, so that every subarray it considers is of the form A[i..mid]. Lines 1–2 initialize the variables left-sum, which holds the greatest sum found so far, and sum, holding the sum of the entries in A[i..mid]. Whenever we find, in line 5, a subarray A[i..mid] with a sum of values greater than left-sum, we update left-sum to this subarray's sum in line 6, and in line 7 we update the variable max-left to record this index *i*. Lines 8–14 work analogously for the right half, A[mid+1..high]. Here, the **for** loop of lines 10–14 starts the index *j* at mid+1 and works up to high, so that every subarray it considers is of the form A[mid + 1..j]. Finally, line 15 returns the indices max-left and max-right that demarcate a maximum subarray crossing the midpoint, along with the sum left-sum + right-sum of the values in the subarray A[max-left..max-right].

If the subarray A[low..high] contains *n* entries (so that n = high - low + 1), we claim that the call FIND-MAX-CROSSING-SUBARRAY (*A*, *low*, *mid*, *high*) takes $\Theta(n)$ time. Since each iteration of each of the two **for** loops takes $\Theta(1)$ time, we just need to count up how many iterations there are altogether. The **for** loop of lines 3–7 makes *mid* – *low* + 1 iterations, and the **for** loop of lines 10–14 makes *high* – *mid* iterations, and so the total number of iterations is

(mid - low + 1) + (high - mid) = high - low + 1= n.

With a linear-time FIND-MAX-CROSSING-SUBARRAY procedure in hand, we can write pseudocode for a divide-and-conquer algorithm to solve the maximum-subarray problem:

FIND-MAXIMUM-SUBARRAY (A, low, high)

1 **if** high == low

5

8

- 2 return (low, high, A[low]) // base case: only one element
- 3 else $mid = \lfloor (low + high)/2 \rfloor$

4 (left-low, left-high, left-sum) =FIND-MAXIMUM-SUBARRAY(A, low, mid)

- (right-low, right-high, right-sum) =FIND-MAXIMUM-SUBARRAY(A, mid + 1, high)
- 6 (cross-low, cross-high, cross-sum) =
 FIND-MAX-CROSSING-SUBARRAY (A, low, mid, high)
 7 if left-sum > right-sum and left-sum > cross-sum
 - **return** (*left-low*, *left-high*, *left-sum*)
- 9 **elseif** right-sum > left-sum and right-sum > cross-sum
- 10 **return** (*right-low*, *right-high*, *right-sum*)
- 11 **else return** (*cross-low*, *cross-high*, *cross-sum*)

The initial call FIND-MAXIMUM-SUBARRAY (A, 1, A.length) will find a maximum subarray of A[1..n].

Similar to FIND-MAX-CROSSING-SUBARRAY, the recursive procedure FIND-MAXIMUM-SUBARRAY returns a tuple containing the indices that demarcate a maximum subarray, along with the sum of the values in a maximum subarray. Line 1 tests for the base case, where the subarray has just one element. A subarray with just one element has only one subarray—itself—and so line 2 returns a tuple with the starting and ending indices of just the one element, along with its value. Lines 3–11 handle the recursive case. Line 3 does the divide part, computing the index *mid* of the midpoint. Let's refer to the subarray A[low..mid] as the *left subarray* and to A[mid + 1..high] as the *right subarray*. Because we know that the subarray A[low ... high] contains at least two elements, each of the left and right subarrays must have at least one element. Lines 4 and 5 conquer by recursively finding maximum subarrays within the left and right subarrays, respectively. Lines 6–11 form the combine part. Line 6 finds a maximum subarray that crosses the midpoint. (Recall that because line 6 solves a subproblem that is not a smaller instance of the original problem, we consider it to be in the combine part.) Line 7 tests whether the left subarray contains a subarray with the maximum sum, and line 8 returns that maximum subarray. Otherwise, line 9 tests whether the right subarray contains a subarray with the maximum sum, and line 10 returns that maximum subarray. If neither the left nor right subarrays contain a subarray achieving the maximum sum, then a maximum subarray must cross the midpoint, and line 11 returns it.

Analyzing the divide-and-conquer algorithm

Next we set up a recurrence that describes the running time of the recursive FIND-MAXIMUM-SUBARRAY procedure. As we did when we analyzed merge sort in Section 2.3.2, we make the simplifying assumption that the original problem size is a power of 2, so that all subproblem sizes are integers. We denote by T(n) the running time of FIND-MAXIMUM-SUBARRAY on a subarray of n elements. For starters, line 1 takes constant time. The base case, when n = 1, is easy: line 2 takes constant time, and so

$$T(1) = \Theta(1) . \tag{4.5}$$

The recursive case occurs when n > 1. Lines 1 and 3 take constant time. Each of the subproblems solved in lines 4 and 5 is on a subarray of n/2 elements (our assumption that the original problem size is a power of 2 ensures that n/2 is an integer), and so we spend T(n/2) time solving each of them. Because we have to solve two subproblems—for the left subarray and for the right subarray—the contribution to the running time from lines 4 and 5 comes to 2T(n/2). As we have

already seen, the call to FIND-MAX-CROSSING-SUBARRAY in line 6 takes $\Theta(n)$ time. Lines 7–11 take only $\Theta(1)$ time. For the recursive case, therefore, we have

$$T(n) = \Theta(1) + 2T(n/2) + \Theta(n) + \Theta(1) = 2T(n/2) + \Theta(n) .$$
(4.6)

Combining equations (4.5) and (4.6) gives us a recurrence for the running time T(n) of FIND-MAXIMUM-SUBARRAY:

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ 2T(n/2) + \Theta(n) & \text{if } n > 1 . \end{cases}$$
(4.7)

This recurrence is the same as recurrence (4.1) for merge sort. As we shall see from the master method in Section 4.5, this recurrence has the solution $T(n) = \Theta(n \lg n)$. You might also revisit the recursion tree in Figure 2.5 to understand why the solution should be $T(n) = \Theta(n \lg n)$.

Thus, we see that the divide-and-conquer method yields an algorithm that is asymptotically faster than the brute-force method. With merge sort and now the maximum-subarray problem, we begin to get an idea of how powerful the divideand-conquer method can be. Sometimes it will yield the asymptotically fastest algorithm for a problem, and other times we can do even better. As Exercise 4.1-5 shows, there is in fact a linear-time algorithm for the maximum-subarray problem, and it does not use divide-and-conquer.

Exercises

4.1-1

What does FIND-MAXIMUM-SUBARRAY return when all elements of *A* are negative?

4.1-2

Write pseudocode for the brute-force method of solving the maximum-subarray problem. Your procedure should run in $\Theta(n^2)$ time.

4.1-3

Implement both the brute-force and recursive algorithms for the maximumsubarray problem on your own computer. What problem size n_0 gives the crossover point at which the recursive algorithm beats the brute-force algorithm? Then, change the base case of the recursive algorithm to use the brute-force algorithm whenever the problem size is less than n_0 . Does that change the crossover point?

4.1-4

Suppose we change the definition of the maximum-subarray problem to allow the result to be an empty subarray, where the sum of the values of an empty subar-

ray is 0. How would you change any of the algorithms that do not allow empty subarrays to permit an empty subarray to be the result?

4.1-5

Use the following ideas to develop a nonrecursive, linear-time algorithm for the maximum-subarray problem. Start at the left end of the array, and progress toward the right, keeping track of the maximum subarray seen so far. Knowing a maximum subarray of A[1.. j], extend the answer to find a maximum subarray ending at index j + 1 by using the following observation: a maximum subarray of A[1.. j + 1] is either a maximum subarray of A[1.. j] or a subarray A[i.. j + 1], for some $1 \le i \le j + 1$. Determine a maximum subarray of the form A[i.. j + 1] in constant time based on knowing a maximum subarray ending at index j.

4.2 Strassen's algorithm for matrix multiplication

If you have seen matrices before, then you probably know how to multiply them. (Otherwise, you should read Section D.1 in Appendix D.) If $A = (a_{ij})$ and $B = (b_{ij})$ are square $n \times n$ matrices, then in the product $C = A \cdot B$, we define the entry c_{ij} , for i, j = 1, 2, ..., n, by

$$c_{ij} = \sum_{k=1}^{n} a_{ik} \cdot b_{kj} .$$
(4.8)

We must compute n^2 matrix entries, and each is the sum of *n* values. The following procedure takes $n \times n$ matrices *A* and *B* and multiplies them, returning their $n \times n$ product *C*. We assume that each matrix has an attribute *rows*, giving the number of rows in the matrix.

SQUARE-MATRIX-MULTIPLY (A, B)

```
1 n = A.rows

2 let C be a new n \times n matrix

3 for i = 1 to n

4 for j = 1 to n

5 c_{ij} = 0

6 for k = 1 to n

7 c_{ij} = c_{ij} + a_{ik} \cdot b_{kj}

8 return C
```

The SQUARE-MATRIX-MULTIPLY procedure works as follows. The **for** loop of lines 3-7 computes the entries of each row *i*, and within a given row *i*, the

for loop of lines 4–7 computes each of the entries c_{ij} , for each column *j*. Line 5 initializes c_{ij} to 0 as we start computing the sum given in equation (4.8), and each iteration of the **for** loop of lines 6–7 adds in one more term of equation (4.8).

Because each of the triply-nested **for** loops runs exactly *n* iterations, and each execution of line 7 takes constant time, the SQUARE-MATRIX-MULTIPLY procedure takes $\Theta(n^3)$ time.

You might at first think that any matrix multiplication algorithm must take $\Omega(n^3)$ time, since the natural definition of matrix multiplication requires that many multiplications. You would be incorrect, however: we have a way to multiply matrices in $o(n^3)$ time. In this section, we shall see Strassen's remarkable recursive algorithm for multiplying $n \times n$ matrices. It runs in $\Theta(n^{\lg 7})$ time, which we shall show in Section 4.5. Since lg 7 lies between 2.80 and 2.81, Strassen's algorithm runs in $O(n^{2.81})$ time, which is asymptotically better than the simple SQUARE-MATRIX-MULTIPLY procedure.

A simple divide-and-conquer algorithm

To keep things simple, when we use a divide-and-conquer algorithm to compute the matrix product $C = A \cdot B$, we assume that *n* is an exact power of 2 in each of the $n \times n$ matrices. We make this assumption because in each divide step, we will divide $n \times n$ matrices into four $n/2 \times n/2$ matrices, and by assuming that *n* is an exact power of 2, we are guaranteed that as long as $n \ge 2$, the dimension n/2 is an integer.

Suppose that we partition each of A, B, and C into four $n/2 \times n/2$ matrices

$$A = \begin{pmatrix} A_{11} & A_{12} \\ A_{21} & A_{22} \end{pmatrix}, \quad B = \begin{pmatrix} B_{11} & B_{12} \\ B_{21} & B_{22} \end{pmatrix}, \quad C = \begin{pmatrix} C_{11} & C_{12} \\ C_{21} & C_{22} \end{pmatrix}, \quad (4.9)$$

so that we rewrite the equation $C = A \cdot B$ as

$$\begin{pmatrix} C_{11} & C_{12} \\ C_{21} & C_{22} \end{pmatrix} = \begin{pmatrix} A_{11} & A_{12} \\ A_{21} & A_{22} \end{pmatrix} \cdot \begin{pmatrix} B_{11} & B_{12} \\ B_{21} & B_{22} \end{pmatrix}.$$
 (4.10)

Equation (4.10) corresponds to the four equations

$$C_{11} = A_{11} \cdot B_{11} + A_{12} \cdot B_{21} , \qquad (4.11)$$

$$C_{12} = A_{11} \cdot B_{12} + A_{12} \cdot B_{22} , \qquad (4.12)$$

$$C_{21} = A_{21} \cdot B_{11} + A_{22} \cdot B_{21} , \qquad (4.13)$$

$$C_{22} = A_{21} \cdot B_{12} + A_{22} \cdot B_{22} . (4.14)$$

Each of these four equations specifies two multiplications of $n/2 \times n/2$ matrices and the addition of their $n/2 \times n/2$ products. We can use these equations to create a straightforward, recursive, divide-and-conquer algorithm:

SQUARE-MATRIX-MULTIPLY-RECURSIVE(A, B)

1	n = A.rows
2	let C be a new $n \times n$ matrix
3	if $n == 1$
4	$c_{11} = a_{11} \cdot b_{11}$
5	else partition A, B, and C as in equations (4.9)
6	$C_{11} = $ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{11}, B_{11})
	+ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{12}, B_{21})
7	$C_{12} = $ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{11}, B_{12})
	+ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{12}, B_{22})
8	$C_{21} = $ Square-Matrix-Multiply-Recursive (A_{21}, B_{11})
	+ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{22}, B_{21})
9	$C_{22} = $ Square-Matrix-Multiply-Recursive (A_{21}, B_{12})
	+ SQUARE-MATRIX-MULTIPLY-RECURSIVE (A_{22}, B_{22})
10	

10 return C

This pseudocode glosses over one subtle but important implementation detail. How do we partition the matrices in line 5? If we were to create 12 new $n/2 \times n/2$ matrices, we would spend $\Theta(n^2)$ time copying entries. In fact, we can partition the matrices without copying entries. The trick is to use index calculations. We identify a submatrix by a range of row indices and a range of column indices of the original matrix. We end up representing a submatrix a little differently from how we represent the original matrix, which is the subtlety we are glossing over. The advantage is that, since we can specify submatrices by index calculations, executing line 5 takes only $\Theta(1)$ time (although we shall see that it makes no difference asymptotically to the overall running time whether we copy or partition in place).

Now, we derive a recurrence to characterize the running time of SQUARE-MATRIX-MULTIPLY-RECURSIVE. Let T(n) be the time to multiply two $n \times n$ matrices using this procedure. In the base case, when n = 1, we perform just the one scalar multiplication in line 4, and so

$$T(1) = \Theta(1)$$
. (4.15)

The recursive case occurs when n > 1. As discussed, partitioning the matrices in line 5 takes $\Theta(1)$ time, using index calculations. In lines 6–9, we recursively call SQUARE-MATRIX-MULTIPLY-RECURSIVE a total of eight times. Because each recursive call multiplies two $n/2 \times n/2$ matrices, thereby contributing T(n/2) to the overall running time, the time taken by all eight recursive calls is 8T(n/2). We also must account for the four matrix additions in lines 6–9. Each of these matrices contains $n^2/4$ entries, and so each of the four matrix additions takes $\Theta(n^2)$ time. Since the number of matrix additions is a constant, the total time spent adding ma-

trices in lines 6–9 is $\Theta(n^2)$. (Again, we use index calculations to place the results of the matrix additions into the correct positions of matrix *C*, with an overhead of $\Theta(1)$ time per entry.) The total time for the recursive case, therefore, is the sum of the partitioning time, the time for all the recursive calls, and the time to add the matrices resulting from the recursive calls:

$$T(n) = \Theta(1) + 8T(n/2) + \Theta(n^2) = 8T(n/2) + \Theta(n^2) .$$
(4.16)

Notice that if we implemented partitioning by copying matrices, which would cost $\Theta(n^2)$ time, the recurrence would not change, and hence the overall running time would increase by only a constant factor.

Combining equations (4.15) and (4.16) gives us the recurrence for the running time of SQUARE-MATRIX-MULTIPLY-RECURSIVE:

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ 8T(n/2) + \Theta(n^2) & \text{if } n > 1 . \end{cases}$$
(4.17)

As we shall see from the master method in Section 4.5, recurrence (4.17) has the solution $T(n) = \Theta(n^3)$. Thus, this simple divide-and-conquer approach is no faster than the straightforward SQUARE-MATRIX-MULTIPLY procedure.

Before we continue on to examining Strassen's algorithm, let us review where the components of equation (4.16) came from. Partitioning each $n \times n$ matrix by index calculation takes $\Theta(1)$ time, but we have two matrices to partition. Although you could say that partitioning the two matrices takes $\Theta(2)$ time, the constant of 2 is subsumed by the Θ -notation. Adding two matrices, each with, say, k entries, takes $\Theta(k)$ time. Since the matrices we add each have $n^2/4$ entries, you could say that adding each pair takes $\Theta(n^2/4)$ time. Again, however, the Θ -notation subsumes the constant factor of 1/4, and we say that adding two $n^2/4 \times n^2/4$ matrices takes $\Theta(n^2)$ time. We have four such matrix additions, and once again, instead of saying that they take $\Theta(4n^2)$ time, we say that they take $\Theta(n^2)$ time. (Of course, you might observe that we could say that the four matrix additions take $\Theta(4n^2/4)$ time, and that $4n^2/4 = n^2$, but the point here is that Θ -notation subsumes constant factors, whatever they are.) Thus, we end up with two terms of $\Theta(n^2)$, which we can combine into one.

When we account for the eight recursive calls, however, we cannot just subsume the constant factor of 8. In other words, we must say that together they take 8T(n/2) time, rather than just T(n/2) time. You can get a feel for why by looking back at the recursion tree in Figure 2.5, for recurrence (2.1) (which is identical to recurrence (4.7)), with the recursive case $T(n) = 2T(n/2) + \Theta(n)$. The factor of 2 determined how many children each tree node had, which in turn determined how many terms contributed to the sum at each level of the tree. If we were to ignore



the factor of 8 in equation (4.16) or the factor of 2 in recurrence (4.1), the recursion tree would just be linear, rather than "bushy," and each level would contribute only one term to the sum.

Bear in mind, therefore, that although asymptotic notation subsumes constant multiplicative factors, recursive notation such as T(n/2) does not.

Strassen's method

The key to Strassen's method is to make the recursion tree slightly less bushy. That is, instead of performing eight recursive multiplications of $n/2 \times n/2$ matrices, it performs only seven. The cost of eliminating one matrix multiplication will be several new additions of $n/2 \times n/2$ matrices, but still only a constant number of additions. As before, the constant number of matrix additions will be subsumed by Θ -notation when we set up the recurrence equation to characterize the running time.

Strassen's method is not at all obvious. (This might be the biggest understatement in this book.) It has four steps:

- 1. Divide the input matrices A and B and output matrix C into $n/2 \times n/2$ submatrices, as in equation (4.9). This step takes $\Theta(1)$ time by index calculation, just as in SQUARE-MATRIX-MULTIPLY-RECURSIVE.
- 2. Create 10 matrices S_1, S_2, \ldots, S_{10} , each of which is $n/2 \times n/2$ and is the sum or difference of two matrices created in step 1. We can create all 10 matrices in $\Theta(n^2)$ time.
- 3. Using the submatrices created in step 1 and the 10 matrices created in step 2, recursively compute seven matrix products P_1, P_2, \ldots, P_7 . Each matrix P_i is $n/2 \times n/2$.
- 4. Compute the desired submatrices $C_{11}, C_{12}, C_{21}, C_{22}$ of the result matrix C by adding and subtracting various combinations of the P_i matrices. We can compute all four submatrices in $\Theta(n^2)$ time.

We shall see the details of steps 2–4 in a moment, but we already have enough information to set up a recurrence for the running time of Strassen's method. Let us assume that once the matrix size *n* gets down to 1, we perform a simple scalar multiplication, just as in line 4 of SQUARE-MATRIX-MULTIPLY-RECURSIVE. When n > 1, steps 1, 2, and 4 take a total of $\Theta(n^2)$ time, and step 3 requires us to perform seven multiplications of $n/2 \times n/2$ matrices. Hence, we obtain the following recurrence for the running time T(n) of Strassen's algorithm:

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1 ,\\ 7T(n/2) + \Theta(n^2) & \text{if } n > 1 . \end{cases}$$
(4.18)

We have traded off one matrix multiplication for a constant number of matrix additions. Once we understand recurrences and their solutions, we shall see that this tradeoff actually leads to a lower asymptotic running time. By the master method in Section 4.5, recurrence (4.18) has the solution $T(n) = \Theta(n^{\lg 7})$.

We now proceed to describe the details. In step 2, we create the following 10 matrices:

$$S_{1} = B_{12} - B_{22} ,$$

$$S_{2} = A_{11} + A_{12} ,$$

$$S_{3} = A_{21} + A_{22} ,$$

$$S_{4} = B_{21} - B_{11} ,$$

$$S_{5} = A_{11} + A_{22} ,$$

$$S_{6} = B_{11} + B_{22} ,$$

$$S_{7} = A_{12} - A_{22} ,$$

$$S_{8} = B_{21} + B_{22} ,$$

$$S_{9} = A_{11} - A_{21} ,$$

$$S_{10} = B_{11} + B_{12} .$$

Since we must add or subtract $n/2 \times n/2$ matrices 10 times, this step does indeed take $\Theta(n^2)$ time.

In step 3, we recursively multiply $n/2 \times n/2$ matrices seven times to compute the following $n/2 \times n/2$ matrices, each of which is the sum or difference of products of *A* and *B* submatrices:

$$P_{1} = A_{11} \cdot S_{1} = A_{11} \cdot B_{12} - A_{11} \cdot B_{22} ,$$

$$P_{2} = S_{2} \cdot B_{22} = A_{11} \cdot B_{22} + A_{12} \cdot B_{22} ,$$

$$P_{3} = S_{3} \cdot B_{11} = A_{21} \cdot B_{11} + A_{22} \cdot B_{11} ,$$

$$P_{4} = A_{22} \cdot S_{4} = A_{22} \cdot B_{21} - A_{22} \cdot B_{11} ,$$

$$P_{5} = S_{5} \cdot S_{6} = A_{11} \cdot B_{11} + A_{11} \cdot B_{22} + A_{22} \cdot B_{11} + A_{22} \cdot B_{22} ,$$

$$P_{6} = S_{7} \cdot S_{8} = A_{12} \cdot B_{21} + A_{12} \cdot B_{22} - A_{22} \cdot B_{21} - A_{22} \cdot B_{22} ,$$

$$P_{7} = S_{9} \cdot S_{10} = A_{11} \cdot B_{11} + A_{11} \cdot B_{12} - A_{21} \cdot B_{11} - A_{21} \cdot B_{12} .$$

Note that the only multiplications we need to perform are those in the middle column of the above equations. The right-hand column just shows what these products equal in terms of the original submatrices created in step 1.

Step 4 adds and subtracts the P_i matrices created in step 3 to construct the four $n/2 \times n/2$ submatrices of the product C. We start with

$$C_{11} = P_5 + P_4 - P_2 + P_6 \, .$$

Expanding out the right-hand side, with the expansion of each P_i on its own line and vertically aligning terms that cancel out, we see that C_{11} equals

$$\begin{array}{c} A_{11} \cdot B_{11} + A_{11} \cdot B_{22} + A_{22} \cdot B_{11} + A_{22} \cdot B_{22} \\ & -A_{22} \cdot B_{11} + A_{22} \cdot B_{21} \\ & -A_{11} \cdot B_{22} - A_{22} \cdot B_{22} - A_{22} \cdot B_{21} + A_{12} \cdot B_{22} + A_{12} \cdot B_{21} \end{array}$$

 $A_{11} \cdot B_{11}$

 $+A_{12} \cdot B_{21}$,

which corresponds to equation (4.11).

Similarly, we set

 $C_{12} = P_1 + P_2$,

and so C_{12} equals

$$\begin{array}{c} A_{11} \cdot B_{12} - A_{11} \cdot B_{22} \\ + A_{11} \cdot B_{22} + A_{12} \cdot B_{22} \end{array}$$

$$A_{11} \cdot B_{12} \qquad \qquad + A_{12} \cdot B_{22}$$

corresponding to equation (4.12).

Setting

$$C_{21} = P_3 + P_4$$

makes C_{21} equal

$$\begin{array}{c} A_{21} \cdot B_{11} + A_{22} \cdot B_{11} \\ - A_{22} \cdot B_{11} + A_{22} \cdot B_{21} \end{array}$$

$$A_{21} \cdot B_{11} + A_{22} \cdot B_{21} ,$$

corresponding to equation (4.13).

Finally, we set

$$C_{22} = P_5 + P_1 - P_3 - P_7 \; ,$$

so that C_{22} equals

$$\begin{array}{cccc} A_{11} \cdot B_{11} + A_{11} \cdot B_{22} + A_{22} \cdot B_{11} + A_{22} \cdot B_{22} \\ & -A_{11} \cdot B_{22} & + A_{11} \cdot B_{12} \\ & -A_{22} \cdot B_{11} & -A_{21} \cdot B_{11} \\ -A_{11} \cdot B_{11} & -A_{11} \cdot B_{12} + A_{21} \cdot B_{11} + A_{21} \cdot B_{12} \end{array}$$

 $A_{22} \cdot B_{22}$

 $+ A_{21} \cdot B_{12}$,

which corresponds to equation (4.14). Altogether, we add or subtract $n/2 \times n/2$ matrices eight times in step 4, and so this step indeed takes $\Theta(n^2)$ time.

Thus, we see that Strassen's algorithm, comprising steps 1–4, produces the correct matrix product and that recurrence (4.18) characterizes its running time. Since we shall see in Section 4.5 that this recurrence has the solution $T(n) = \Theta(n^{\lg 7})$, Strassen's method is asymptotically faster than the straightforward SQUARE-MATRIX-MULTIPLY procedure. The notes at the end of this chapter discuss some of the practical aspects of Strassen's algorithm.

Exercises

Note: Although Exercises 4.2-3, 4.2-4, and 4.2-5 are about variants on Strassen's algorithm, you should read Section 4.5 before trying to solve them.

4.2-1

Use Strassen's algorithm to compute the matrix product

 $\left(\begin{array}{rrr}1 & 3\\7 & 5\end{array}\right)\left(\begin{array}{rrr}6 & 8\\4 & 2\end{array}\right).$

Show your work.

4.2-2

Write pseudocode for Strassen's algorithm.

4.2-3

How would you modify Strassen's algorithm to multiply $n \times n$ matrices in which n is not an exact power of 2? Show that the resulting algorithm runs in time $\Theta(n^{\lg 7})$.

4.2-4

What is the largest k such that if you can multiply 3×3 matrices using k multiplications (not assuming commutativity of multiplication), then you can multiply $n \times n$ matrices in time $o(n^{\lg 7})$? What would the running time of this algorithm be?

4.2-5

V. Pan has discovered a way of multiplying 68×68 matrices using 132,464 multiplications, a way of multiplying 70×70 matrices using 143,640 multiplications, and a way of multiplying 72×72 matrices using 155,424 multiplications. Which method yields the best asymptotic running time when used in a divide-and-conquer matrix-multiplication algorithm? How does it compare to Strassen's algorithm?
4.2-6

How quickly can you multiply a $kn \times n$ matrix by an $n \times kn$ matrix, using Strassen's algorithm as a subroutine? Answer the same question with the order of the input matrices reversed.

4.2-7

Show how to multiply the complex numbers a + bi and c + di using only three multiplications of real numbers. The algorithm should take a, b, c, and d as input and produce the real component ac - bd and the imaginary component ad + bc separately.

4.3 The substitution method for solving recurrences

Now that we have seen how recurrences characterize the running times of divideand-conquer algorithms, we will learn how to solve recurrences. We start in this section with the "substitution" method.

The *substitution method* for solving recurrences comprises two steps:

- 1. Guess the form of the solution.
- 2. Use mathematical induction to find the constants and show that the solution works.

We substitute the guessed solution for the function when applying the inductive hypothesis to smaller values; hence the name "substitution method." This method is powerful, but we must be able to guess the form of the answer in order to apply it.

We can use the substitution method to establish either upper or lower bounds on a recurrence. As an example, let us determine an upper bound on the recurrence

$$T(n) = 2T(\lfloor n/2 \rfloor) + n , \qquad (4.19)$$

which is similar to recurrences (4.3) and (4.4). We guess that the solution is $T(n) = O(n \lg n)$. The substitution method requires us to prove that $T(n) \le cn \lg n$ for an appropriate choice of the constant c > 0. We start by assuming that this bound holds for all positive m < n, in particular for $m = \lfloor n/2 \rfloor$, yielding $T(\lfloor n/2 \rfloor) \le c \lfloor n/2 \rfloor \lg(\lfloor n/2 \rfloor)$. Substituting into the recurrence yields

$$T(n) \leq 2(c \lfloor n/2 \rfloor \lg(\lfloor n/2 \rfloor)) + n$$

- $\leq cn \lg(n/2) + n$
- $= cn \lg n cn \lg 2 + n$
- $= cn \lg n cn + n$
- $\leq cn \lg n$,

where the last step holds as long as $c \ge 1$.

Mathematical induction now requires us to show that our solution holds for the boundary conditions. Typically, we do so by showing that the boundary conditions are suitable as base cases for the inductive proof. For the recurrence (4.19), we must show that we can choose the constant c large enough so that the bound $T(n) \le cn \lg n$ works for the boundary conditions as well. This requirement can sometimes lead to problems. Let us assume, for the sake of argument, that T(1) = 1 is the sole boundary condition of the recurrence. Then for n = 1, the bound $T(n) \le cn \lg n$ yields $T(1) \le c1 \lg 1 = 0$, which is at odds with T(1) = 1. Consequently, the base case of our inductive proof fails to hold.

We can overcome this obstacle in proving an inductive hypothesis for a specific boundary condition with only a little more effort. In the recurrence (4.19), for example, we take advantage of asymptotic notation requiring us only to prove $T(n) \leq cn \lg n$ for $n \geq n_0$, where n_0 is a constant that we get to choose. We keep the troublesome boundary condition T(1) = 1, but remove it from consideration in the inductive proof. We do so by first observing that for n > 3, the recurrence does not depend directly on T(1). Thus, we can replace T(1) by T(2)and T(3) as the base cases in the inductive proof, letting $n_0 = 2$. Note that we make a distinction between the base case of the recurrence (n = 1) and the base cases of the inductive proof (n = 2 and n = 3). With T(1) = 1, we derive from the recurrence that T(2) = 4 and T(3) = 5. Now we can complete the inductive proof that $T(n) < cn \lg n$ for some constant c > 1 by choosing c large enough so that $T(2) \le c2 \lg 2$ and $T(3) \le c3 \lg 3$. As it turns out, any choice of $c \ge 2$ suffices for the base cases of n = 2 and n = 3 to hold. For most of the recurrences we shall examine, it is straightforward to extend boundary conditions to make the inductive assumption work for small n, and we shall not always explicitly work out the details.

Making a good guess

Unfortunately, there is no general way to guess the correct solutions to recurrences. Guessing a solution takes experience and, occasionally, creativity. Fortunately, though, you can use some heuristics to help you become a good guesser. You can also use recursion trees, which we shall see in Section 4.4, to generate good guesses.

If a recurrence is similar to one you have seen before, then guessing a similar solution is reasonable. As an example, consider the recurrence

$$T(n) = 2T(|n/2| + 17) + n$$
,

which looks difficult because of the added "17" in the argument to T on the righthand side. Intuitively, however, this additional term cannot substantially affect the



solution to the recurrence. When *n* is large, the difference between $\lfloor n/2 \rfloor$ and $\lfloor n/2 \rfloor + 17$ is not that large: both cut *n* nearly evenly in half. Consequently, we make the guess that $T(n) = O(n \lg n)$, which you can verify as correct by using the substitution method (see Exercise 4.3-6).

Another way to make a good guess is to prove loose upper and lower bounds on the recurrence and then reduce the range of uncertainty. For example, we might start with a lower bound of $T(n) = \Omega(n)$ for the recurrence (4.19), since we have the term *n* in the recurrence, and we can prove an initial upper bound of $T(n) = O(n^2)$. Then, we can gradually lower the upper bound and raise the lower bound until we converge on the correct, asymptotically tight solution of $T(n) = \Theta(n \lg n)$.

Subtleties

Sometimes you might correctly guess an asymptotic bound on the solution of a recurrence, but somehow the math fails to work out in the induction. The problem frequently turns out to be that the inductive assumption is not strong enough to prove the detailed bound. If you revise the guess by subtracting a lower-order term when you hit such a snag, the math often goes through.

Consider the recurrence

$$T(n) = T(|n/2|) + T([n/2]) + 1$$
.

We guess that the solution is T(n) = O(n), and we try to show that $T(n) \le cn$ for an appropriate choice of the constant *c*. Substituting our guess in the recurrence, we obtain

$$T(n) \leq c \lfloor n/2 \rfloor + c \lceil n/2 \rceil + 1$$

= $cn + 1$,

which does not imply $T(n) \le cn$ for any choice of c. We might be tempted to try a larger guess, say $T(n) = O(n^2)$. Although we can make this larger guess work, our original guess of T(n) = O(n) is correct. In order to show that it is correct, however, we must make a stronger inductive hypothesis.

Intuitively, our guess is nearly right: we are off only by the constant 1, a lower-order term. Nevertheless, mathematical induction does not work unless we prove the exact form of the inductive hypothesis. We overcome our difficulty by *subtracting* a lower-order term from our previous guess. Our new guess is $T(n) \le cn - d$, where $d \ge 0$ is a constant. We now have

$$T(n) \leq (c \lfloor n/2 \rfloor - d) + (c \lceil n/2 \rceil - d) + 1$$

= $cn - 2d + 1$
 $\leq cn - d$,

as long as $d \ge 1$. As before, we must choose the constant *c* large enough to handle the boundary conditions.

You might find the idea of subtracting a lower-order term counterintuitive. After all, if the math does not work out, we should increase our guess, right? Not necessarily! When proving an upper bound by induction, it may actually be more difficult to prove that a weaker upper bound holds, because in order to prove the weaker bound, we must use the same weaker bound inductively in the proof. In our current example, when the recurrence has more than one recursive term, we get to subtract out the lower-order term of the proposed bound once per recursive term. In the above example, we subtracted out the constant *d* twice, once for the $T(\lfloor n/2 \rfloor)$ term and once for the $T(\lfloor n/2 \rceil)$ term. We ended up with the inequality $T(n) \le cn - 2d + 1$, and it was easy to find values of *d* to make cn - 2d + 1 be less than or equal to cn - d.

Avoiding pitfalls

It is easy to err in the use of asymptotic notation. For example, in the recurrence (4.19) we can falsely "prove" T(n) = O(n) by guessing $T(n) \le cn$ and then arguing

$$T(n) \leq 2(c \lfloor n/2 \rfloor) + n$$

$$\leq cn + n$$

$$= O(n), \qquad \longleftarrow wrong!!$$

since *c* is a constant. The error is that we have not proved the *exact form* of the inductive hypothesis, that is, that $T(n) \leq cn$. We therefore will explicitly prove that $T(n) \leq cn$ when we want to show that T(n) = O(n).

Changing variables

Sometimes, a little algebraic manipulation can make an unknown recurrence similar to one you have seen before. As an example, consider the recurrence

$$T(n) = 2T\left(\left\lfloor\sqrt{n}\right\rfloor\right) + \lg n$$
,

which looks difficult. We can simplify this recurrence, though, with a change of variables. For convenience, we shall not worry about rounding off values, such as \sqrt{n} , to be integers. Renaming $m = \lg n$ yields

$$T(2^m) = 2T(2^{m/2}) + m$$
.

We can now rename $S(m) = T(2^m)$ to produce the new recurrence

$$S(m) = 2S(m/2) + m$$
,

which is very much like recurrence (4.19). Indeed, this new recurrence has the same solution: $S(m) = O(m \lg m)$. Changing back from S(m) to T(n), we obtain

$$T(n) = T(2^m) = S(m) = O(m \lg m) = O(\lg n \lg \lg n)$$

Exercises

4.3-1

Show that the solution of T(n) = T(n-1) + n is $O(n^2)$.

4.3-2

Show that the solution of $T(n) = T(\lceil n/2 \rceil) + 1$ is $O(\lg n)$.

4.3-3

We saw that the solution of $T(n) = 2T(\lfloor n/2 \rfloor) + n$ is $O(n \lg n)$. Show that the solution of this recurrence is also $\Omega(n \lg n)$. Conclude that the solution is $\Theta(n \lg n)$.

4.3-4

Show that by making a different inductive hypothesis, we can overcome the difficulty with the boundary condition T(1) = 1 for recurrence (4.19) without adjusting the boundary conditions for the inductive proof.

4.3-5

Show that $\Theta(n \lg n)$ is the solution to the "exact" recurrence (4.3) for merge sort.

4.3-6

Show that the solution to $T(n) = 2T(\lfloor n/2 \rfloor + 17) + n$ is $O(n \lg n)$.

*4.3-*7

Using the master method in Section 4.5, you can show that the solution to the recurrence T(n) = 4T(n/3) + n is $T(n) = \Theta(n^{\log_3 4})$. Show that a substitution proof with the assumption $T(n) \le cn^{\log_3 4}$ fails. Then show how to subtract off a lower-order term to make a substitution proof work.

4.3-8

Using the master method in Section 4.5, you can show that the solution to the recurrence $T(n) = 4T(n/2) + n^2$ is $T(n) = \Theta(n^2)$. Show that a substitution proof with the assumption $T(n) \le cn^2$ fails. Then show how to subtract off a lower-order term to make a substitution proof work.

4.3-9

Solve the recurrence $T(n) = 3T(\sqrt{n}) + \log n$ by making a change of variables. Your solution should be asymptotically tight. Do not worry about whether values are integral.

4.4 The recursion-tree method for solving recurrences

Although you can use the substitution method to provide a succinct proof that a solution to a recurrence is correct, you might have trouble coming up with a good guess. Drawing out a recursion tree, as we did in our analysis of the merge sort recurrence in Section 2.3.2, serves as a straightforward way to devise a good guess. In a *recursion tree*, each node represents the cost of a single subproblem somewhere in the set of recursive function invocations. We sum the costs within each level of the tree to obtain a set of per-level costs, and then we sum all the per-level costs to determine the total cost of all levels of the recursion.

A recursion tree is best used to generate a good guess, which you can then verify by the substitution method. When using a recursion tree to generate a good guess, you can often tolerate a small amount of "sloppiness," since you will be verifying your guess later on. If you are very careful when drawing out a recursion tree and summing the costs, however, you can use a recursion tree as a direct proof of a solution to a recurrence. In this section, we will use recursion trees to generate good guesses, and in Section 4.6, we will use recursion trees directly to prove the theorem that forms the basis of the master method.

For example, let us see how a recursion tree would provide a good guess for the recurrence $T(n) = 3T(\lfloor n/4 \rfloor) + \Theta(n^2)$. We start by focusing on finding an upper bound for the solution. Because we know that floors and ceilings usually do not matter when solving recurrences (here's an example of sloppiness that we can tolerate), we create a recursion tree for the recurrence $T(n) = 3T(n/4) + cn^2$, having written out the implied constant coefficient c > 0.

Figure 4.5 shows how we derive the recursion tree for $T(n) = 3T(n/4) + cn^2$. For convenience, we assume that *n* is an exact power of 4 (another example of tolerable sloppiness) so that all subproblem sizes are integers. Part (a) of the figure shows T(n), which we expand in part (b) into an equivalent tree representing the recurrence. The cn^2 term at the root represents the cost at the top level of recursion, and the three subtrees of the root represent the costs incurred by the subproblems of size n/4. Part (c) shows this process carried one step further by expanding each node with cost T(n/4) from part (b). The cost for each of the three children of the root is $c(n/4)^2$. We continue expanding each node in the tree by breaking it into its constituent parts as determined by the recurrence.



Figure 4.5 Constructing a recursion tree for the recurrence $T(n) = 3T(n/4) + cn^2$. Part (a) shows T(n), which progressively expands in (b)–(d) to form the recursion tree. The fully expanded tree in part (d) has height $\log_4 n$ (it has $\log_4 n + 1$ levels).

Because subproblem sizes decrease by a factor of 4 each time we go down one level, we eventually must reach a boundary condition. How far from the root do we reach one? The subproblem size for a node at depth *i* is $n/4^i$. Thus, the subproblem size hits n = 1 when $n/4^i = 1$ or, equivalently, when $i = \log_4 n$. Thus, the tree has $\log_4 n + 1$ levels (at depths $0, 1, 2, \ldots, \log_4 n$).

Next we determine the cost at each level of the tree. Each level has three times more nodes than the level above, and so the number of nodes at depth *i* is 3^i . Because subproblem sizes reduce by a factor of 4 for each level we go down from the root, each node at depth *i*, for $i = 0, 1, 2, ..., \log_4 n - 1$, has a cost of $c(n/4^i)^2$. Multiplying, we see that the total cost over all nodes at depth *i*, for $i = 0, 1, 2, ..., \log_4 n - 1$, is $3^i c(n/4^i)^2 = (3/16)^i cn^2$. The bottom level, at depth $\log_4 n$, has $3^{\log_4 n} = n^{\log_4 3}$ nodes, each contributing cost T(1), for a total cost of $n^{\log_4 3}T(1)$, which is $\Theta(n^{\log_4 3})$, since we assume that T(1) is a constant.

Now we add up the costs over all levels to determine the cost for the entire tree:

$$T(n) = cn^{2} + \frac{3}{16}cn^{2} + \left(\frac{3}{16}\right)^{2}cn^{2} + \dots + \left(\frac{3}{16}\right)^{\log_{4}n-1}cn^{2} + \Theta(n^{\log_{4}3})$$

$$= \sum_{i=0}^{\log_{4}n-1} \left(\frac{3}{16}\right)^{i}cn^{2} + \Theta(n^{\log_{4}3})$$

$$= \frac{(3/16)^{\log_{4}n} - 1}{(3/16) - 1}cn^{2} + \Theta(n^{\log_{4}3}) \qquad \text{(by equation (A.5))}.$$

This last formula looks somewhat messy until we realize that we can again take advantage of small amounts of sloppiness and use an infinite decreasing geometric series as an upper bound. Backing up one step and applying equation (A.6), we have

$$T(n) = \sum_{i=0}^{\log_4 n-1} \left(\frac{3}{16}\right)^i cn^2 + \Theta(n^{\log_4 3})$$

$$< \sum_{i=0}^{\infty} \left(\frac{3}{16}\right)^i cn^2 + \Theta(n^{\log_4 3})$$

$$= \frac{1}{1-(3/16)} cn^2 + \Theta(n^{\log_4 3})$$

$$= \frac{16}{13} cn^2 + \Theta(n^{\log_4 3})$$

$$= O(n^2) .$$

Thus, we have derived a guess of $T(n) = O(n^2)$ for our original recurrence $T(n) = 3T(\lfloor n/4 \rfloor) + \Theta(n^2)$. In this example, the coefficients of cn^2 form a decreasing geometric series and, by equation (A.6), the sum of these coefficients



Figure 4.6 A recursion tree for the recurrence T(n) = T(n/3) + T(2n/3) + cn.

is bounded from above by the constant 16/13. Since the root's contribution to the total cost is cn^2 , the root contributes a constant fraction of the total cost. In other words, the cost of the root dominates the total cost of the tree.

In fact, if $O(n^2)$ is indeed an upper bound for the recurrence (as we shall verify in a moment), then it must be a tight bound. Why? The first recursive call contributes a cost of $\Theta(n^2)$, and so $\Omega(n^2)$ must be a lower bound for the recurrence.

Now we can use the substitution method to verify that our guess was correct, that is, $T(n) = O(n^2)$ is an upper bound for the recurrence $T(n) = 3T(\lfloor n/4 \rfloor) + \Theta(n^2)$. We want to show that $T(n) \le dn^2$ for some constant d > 0. Using the same constant c > 0 as before, we have

$$T(n) \leq 3T(\lfloor n/4 \rfloor) + cn^{2}$$

$$\leq 3d \lfloor n/4 \rfloor^{2} + cn^{2}$$

$$\leq 3d(n/4)^{2} + cn^{2}$$

$$= \frac{3}{16}dn^{2} + cn^{2}$$

$$\leq dn^{2},$$

where the last step holds as long as $d \ge (16/13)c$.

In another, more intricate, example, Figure 4.6 shows the recursion tree for

$$T(n) = T(n/3) + T(2n/3) + O(n)$$
.

(Again, we omit floor and ceiling functions for simplicity.) As before, we let c represent the constant factor in the O(n) term. When we add the values across the levels of the recursion tree shown in the figure, we get a value of cn for every level.

The longest simple path from the root to a leaf is $n \to (2/3)n \to (2/3)^2 n \to \dots \to 1$. Since $(2/3)^k n = 1$ when $k = \log_{3/2} n$, the height of the tree is $\log_{3/2} n$.

Intuitively, we expect the solution to the recurrence to be at most the number of levels times the cost of each level, or $O(cn \log_{3/2} n) = O(n \lg n)$. Figure 4.6 shows only the top levels of the recursion tree, however, and not every level in the tree contributes a cost of cn. Consider the cost of the leaves. If this recursion tree were a complete binary tree of height $\log_{3/2} n$, there would be $2^{\log_{3/2} n} = n^{\log_{3/2} 2}$ leaves. Since the cost of each leaf is a constant, the total cost of all leaves would then be $\Theta(n^{\log_{3/2} 2})$ which, since $\log_{3/2} 2$ is a constant strictly greater than 1, is $\omega(n \lg n)$. This recursion tree is not a complete binary tree, however, and so it has fewer than $n^{\log_{3/2} 2}$ leaves. Moreover, as we go down from the root, more and more internal nodes are absent. Consequently, levels toward the bottom of the recursion tree contribute less than cn to the total cost. We could work out an accurate accounting of all costs, but remember that we are just trying to come up with a guess to use in the substitution method. Let us tolerate the sloppiness and attempt to show that a guess of $O(n \lg n)$ for the upper bound is correct.

Indeed, we can use the substitution method to verify that $O(n \lg n)$ is an upper bound for the solution to the recurrence. We show that $T(n) \le dn \lg n$, where d is a suitable positive constant. We have

$$T(n) \leq T(n/3) + T(2n/3) + cn$$

$$\leq d(n/3) \lg(n/3) + d(2n/3) \lg(2n/3) + cn$$

$$= (d(n/3) \lg n - d(n/3) \lg 3) + (d(2n/3) \lg n - d(2n/3) \lg(3/2)) + cn$$

$$= dn \lg n - d((n/3) \lg 3 + (2n/3) \lg(3/2)) + cn$$

$$= dn \lg n - d((n/3) \lg 3 + (2n/3) \lg 3 - (2n/3) \lg 2) + cn$$

$$= dn \lg n - dn(\lg 3 - 2/3) + cn$$

$$\leq dn \lg n ,$$

as long as $d \ge c/(\lg 3 - (2/3))$. Thus, we did not need to perform a more accurate accounting of costs in the recursion tree.

Exercises

4.4-1

Use a recursion tree to determine a good asymptotic upper bound on the recurrence T(n) = 3T(|n/2|) + n. Use the substitution method to verify your answer.

4.4-2

Use a recursion tree to determine a good asymptotic upper bound on the recurrence $T(n) = T(n/2) + n^2$. Use the substitution method to verify your answer.

4.4-3

Use a recursion tree to determine a good asymptotic upper bound on the recurrence T(n) = 4T(n/2 + 2) + n. Use the substitution method to verify your answer.

4.4-4

Use a recursion tree to determine a good asymptotic upper bound on the recurrence T(n) = 2T(n-1) + 1. Use the substitution method to verify your answer.

4.4-5

Use a recursion tree to determine a good asymptotic upper bound on the recurrence T(n) = T(n-1) + T(n/2) + n. Use the substitution method to verify your answer.

4.4-6

Argue that the solution to the recurrence T(n) = T(n/3) + T(2n/3) + cn, where *c* is a constant, is $\Omega(n \lg n)$ by appealing to a recursion tree.

4.4-7

Draw the recursion tree for $T(n) = 4T(\lfloor n/2 \rfloor) + cn$, where c is a constant, and provide a tight asymptotic bound on its solution. Verify your bound by the substitution method.

4.4-8

Use a recursion tree to give an asymptotically tight solution to the recurrence T(n) = T(n-a) + T(a) + cn, where $a \ge 1$ and c > 0 are constants.

4.4-9

Use a recursion tree to give an asymptotically tight solution to the recurrence $T(n) = T(\alpha n) + T((1 - \alpha)n) + cn$, where α is a constant in the range $0 < \alpha < 1$ and c > 0 is also a constant.

4.5 The master method for solving recurrences

The master method provides a "cookbook" method for solving recurrences of the form

$$T(n) = aT(n/b) + f(n),$$
 (4.20)

where $a \ge 1$ and b > 1 are constants and f(n) is an asymptotically positive function. To use the master method, you will need to memorize three cases, but then you will be able to solve many recurrences quite easily, often without pencil and paper.

The recurrence (4.20) describes the running time of an algorithm that divides a problem of size *n* into *a* subproblems, each of size n/b, where *a* and *b* are positive constants. The *a* subproblems are solved recursively, each in time T(n/b). The function f(n) encompasses the cost of dividing the problem and combining the results of the subproblems. For example, the recurrence arising from Strassen's algorithm has a = 7, b = 2, and $f(n) = \Theta(n^2)$.

As a matter of technical correctness, the recurrence is not actually well defined, because n/b might not be an integer. Replacing each of the *a* terms T(n/b) with either $T(\lfloor n/b \rfloor)$ or $T(\lceil n/b \rceil)$ will not affect the asymptotic behavior of the recurrence, however. (We will prove this assertion in the next section.) We normally find it convenient, therefore, to omit the floor and ceiling functions when writing divide-and-conquer recurrences of this form.

The master theorem

The master method depends on the following theorem.

Theorem 4.1 (Master theorem)

Let $a \ge 1$ and b > 1 be constants, let f(n) be a function, and let T(n) be defined on the nonnegative integers by the recurrence

$$T(n) = aT(n/b) + f(n) ,$$

where we interpret n/b to mean either $\lfloor n/b \rfloor$ or $\lceil n/b \rceil$. Then T(n) has the following asymptotic bounds:

- 1. If $f(n) = O(n^{\log_b a \epsilon})$ for some constant $\epsilon > 0$, then $T(n) = \Theta(n^{\log_b a})$.
- 2. If $f(n) = \Theta(n^{\log_b a})$, then $T(n) = \Theta(n^{\log_b a} \lg n)$.
- 3. If $f(n) = \Omega(n^{\log_b a + \epsilon})$ for some constant $\epsilon > 0$, and if $af(n/b) \le cf(n)$ for some constant c < 1 and all sufficiently large n, then $T(n) = \Theta(f(n))$.

Before applying the master theorem to some examples, let's spend a moment trying to understand what it says. In each of the three cases, we compare the function f(n) with the function $n^{\log_b a}$. Intuitively, the larger of the two functions determines the solution to the recurrence. If, as in case 1, the function $n^{\log_b a}$ is the larger, then the solution is $T(n) = \Theta(n^{\log_b a})$. If, as in case 3, the function f(n) is the larger, then the solution is $T(n) = \Theta(f(n))$. If, as in case 2, the two functions are the same size, we multiply by a logarithmic factor, and the solution is $T(n) = \Theta(f(n) \lg n)$.

Beyond this intuition, you need to be aware of some technicalities. In the first case, not only must f(n) be smaller than $n^{\log_b a}$, it must be *polynomially* smaller.

That is, f(n) must be asymptotically smaller than $n^{\log_b a}$ by a factor of n^{ϵ} for some constant $\epsilon > 0$. In the third case, not only must f(n) be larger than $n^{\log_b a}$, it also must be polynomially larger and in addition satisfy the "regularity" condition that $af(n/b) \le cf(n)$. This condition is satisfied by most of the polynomially bounded functions that we shall encounter.

Note that the three cases do not cover all the possibilities for f(n). There is a gap between cases 1 and 2 when f(n) is smaller than $n^{\log_b a}$ but not polynomially smaller. Similarly, there is a gap between cases 2 and 3 when f(n) is larger than $n^{\log_b a}$ but not polynomially larger. If the function f(n) falls into one of these gaps, or if the regularity condition in case 3 fails to hold, you cannot use the master method to solve the recurrence.

Using the master method

To use the master method, we simply determine which case (if any) of the master theorem applies and write down the answer.

As a first example, consider

$$T(n) = 9T(n/3) + n$$

For this recurrence, we have a = 9, b = 3, f(n) = n, and thus we have that $n^{\log_b a} = n^{\log_3 9} = \Theta(n^2)$. Since $f(n) = O(n^{\log_3 9 - \epsilon})$, where $\epsilon = 1$, we can apply case 1 of the master theorem and conclude that the solution is $T(n) = \Theta(n^2)$.

Now consider

$$T(n) = T(2n/3) + 1,$$

in which a = 1, b = 3/2, f(n) = 1, and $n^{\log_b a} = n^{\log_{3/2} 1} = n^0 = 1$. Case 2 applies, since $f(n) = \Theta(n^{\log_b a}) = \Theta(1)$, and thus the solution to the recurrence is $T(n) = \Theta(\lg n)$.

For the recurrence

$$T(n) = 3T(n/4) + n \lg n ,$$

we have a = 3, b = 4, $f(n) = n \lg n$, and $n^{\log_b a} = n^{\log_4 3} = O(n^{0.793})$. Since $f(n) = \Omega(n^{\log_4 3 + \epsilon})$, where $\epsilon \approx 0.2$, case 3 applies if we can show that the regularity condition holds for f(n). For sufficiently large n, we have that $af(n/b) = 3(n/4) \lg(n/4) \le (3/4)n \lg n = cf(n)$ for c = 3/4. Consequently, by case 3, the solution to the recurrence is $T(n) = \Theta(n \lg n)$.

The master method does not apply to the recurrence

$$T(n) = 2T(n/2) + n \lg n ,$$

even though it appears to have the proper form: $a = 2, b = 2, f(n) = n \lg n$, and $n^{\log_b a} = n$. You might mistakenly think that case 3 should apply, since



 $f(n) = n \lg n$ is asymptotically larger than $n^{\log_b a} = n$. The problem is that it is not *polynomially* larger. The ratio $f(n)/n^{\log_b a} = (n \lg n)/n = \lg n$ is asymptotically less than n^{ϵ} for any positive constant ϵ . Consequently, the recurrence falls into the gap between case 2 and case 3. (See Exercise 4.6-2 for a solution.)

Let's use the master method to solve the recurrences we saw in Sections 4.1 and 4.2. Recurrence (4.7),

$$T(n) = 2T(n/2) + \Theta(n) ,$$

characterizes the running times of the divide-and-conquer algorithm for both the maximum-subarray problem and merge sort. (As is our practice, we omit stating the base case in the recurrence.) Here, we have a = 2, b = 2, $f(n) = \Theta(n)$, and thus we have that $n^{\log_b a} = n^{\log_2 2} = n$. Case 2 applies, since $f(n) = \Theta(n)$, and so we have the solution $T(n) = \Theta(n \lg n)$.

Recurrence (4.17),

$$T(n) = 8T(n/2) + \Theta(n^2) ,$$

describes the running time of the first divide-and-conquer algorithm that we saw for matrix multiplication. Now we have a = 8, b = 2, and $f(n) = \Theta(n^2)$, and so $n^{\log_b a} = n^{\log_2 8} = n^3$. Since n^3 is polynomially larger than f(n) (that is, $f(n) = O(n^{3-\epsilon})$ for $\epsilon = 1$), case 1 applies, and $T(n) = \Theta(n^3)$.

Finally, consider recurrence (4.18),

$$T(n) = 7T(n/2) + \Theta(n^2) ,$$

which describes the running time of Strassen's algorithm. Here, we have a = 7, b = 2, $f(n) = \Theta(n^2)$, and thus $n^{\log_b a} = n^{\log_2 7}$. Rewriting $\log_2 7$ as $\lg 7$ and recalling that 2.80 < $\lg 7$ < 2.81, we see that $f(n) = O(n^{\lg 7-\epsilon})$ for $\epsilon = 0.8$. Again, case 1 applies, and we have the solution $T(n) = \Theta(n^{\lg 7})$.

Exercises

4.5-1

Use the master method to give tight asymptotic bounds for the following recurrences.

- a. T(n) = 2T(n/4) + 1.
- **b.** $T(n) = 2T(n/4) + \sqrt{n}$.
- c. T(n) = 2T(n/4) + n.
- **d.** $T(n) = 2T(n/4) + n^2$.

4.5-2

Professor Caesar wishes to develop a matrix-multiplication algorithm that is asymptotically faster than Strassen's algorithm. His algorithm will use the divideand-conquer method, dividing each matrix into pieces of size $n/4 \times n/4$, and the divide and combine steps together will take $\Theta(n^2)$ time. He needs to determine how many subproblems his algorithm has to create in order to beat Strassen's algorithm. If his algorithm creates *a* subproblems, then the recurrence for the running time T(n) becomes $T(n) = aT(n/4) + \Theta(n^2)$. What is the largest integer value of *a* for which Professor Caesar's algorithm would be asymptotically faster than Strassen's algorithm?

4.5-3

Use the master method to show that the solution to the binary-search recurrence $T(n) = T(n/2) + \Theta(1)$ is $T(n) = \Theta(\lg n)$. (See Exercise 2.3-5 for a description of binary search.)

4.5-4

Can the master method be applied to the recurrence $T(n) = 4T(n/2) + n^2 \lg n$? Why or why not? Give an asymptotic upper bound for this recurrence.

4.5-5 *

Consider the regularity condition $af(n/b) \leq cf(n)$ for some constant c < 1, which is part of case 3 of the master theorem. Give an example of constants $a \geq 1$ and b > 1 and a function f(n) that satisfies all the conditions in case 3 of the master theorem except the regularity condition.

★ 4.6 Proof of the master theorem

This section contains a proof of the master theorem (Theorem 4.1). You do not need to understand the proof in order to apply the master theorem.

The proof appears in two parts. The first part analyzes the master recurrence (4.20), under the simplifying assumption that T(n) is defined only on exact powers of b > 1, that is, for $n = 1, b, b^2, \ldots$. This part gives all the intuition needed to understand why the master theorem is true. The second part shows how to extend the analysis to all positive integers n; it applies mathematical technique to the problem of handling floors and ceilings.

In this section, we shall sometimes abuse our asymptotic notation slightly by using it to describe the behavior of functions that are defined only over exact powers of b. Recall that the definitions of asymptotic notations require that

bounds be proved for all sufficiently large numbers, not just those that are powers of b. Since we could make new asymptotic notations that apply only to the set $\{b^i : i = 0, 1, 2, ...\}$, instead of to the nonnegative numbers, this abuse is minor.

Nevertheless, we must always be on guard when we use asymptotic notation over a limited domain lest we draw improper conclusions. For example, proving that T(n) = O(n) when n is an exact power of 2 does not guarantee that T(n) = O(n). The function T(n) could be defined as

$$T(n) = \begin{cases} n & \text{if } n = 1, 2, 4, 8, \dots, \\ n^2 & \text{otherwise}, \end{cases}$$

in which case the best upper bound that applies to all values of *n* is $T(n) = O(n^2)$. Because of this sort of drastic consequence, we shall never use asymptotic notation over a limited domain without making it absolutely clear from the context that we are doing so.

4.6.1 The proof for exact powers

The first part of the proof of the master theorem analyzes the recurrence (4.20)

$$T(n) = aT(n/b) + f(n)$$

for the master method, under the assumption that n is an exact power of b > 1, where b need not be an integer. We break the analysis into three lemmas. The first reduces the problem of solving the master recurrence to the problem of evaluating an expression that contains a summation. The second determines bounds on this summation. The third lemma puts the first two together to prove a version of the master theorem for the case in which n is an exact power of b.

Lemma 4.2

Let $a \ge 1$ and b > 1 be constants, and let f(n) be a nonnegative function defined on exact powers of b. Define T(n) on exact powers of b by the recurrence

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1, \\ aT(n/b) + f(n) & \text{if } n = b^i \end{cases}$$

where i is a positive integer. Then

$$T(n) = \Theta(n^{\log_b a}) + \sum_{j=0}^{\log_b n-1} a^j f(n/b^j) .$$
(4.21)

Proof We use the recursion tree in Figure 4.7. The root of the tree has cost f(n), and it has a children, each with cost f(n/b). (It is convenient to think of a as being



Figure 4.7 The recursion tree generated by T(n) = aT(n/b) + f(n). The tree is a complete *a*-ary tree with $n^{\log_b a}$ leaves and height $\log_b n$. The cost of the nodes at each depth is shown at the right, and their sum is given in equation (4.21).

an integer, especially when visualizing the recursion tree, but the mathematics does not require it.) Each of these children has *a* children, making a^2 nodes at depth 2, and each of the *a* children has cost $f(n/b^2)$. In general, there are a^j nodes at depth *j*, and each has cost $f(n/b^j)$. The cost of each leaf is $T(1) = \Theta(1)$, and each leaf is at depth $\log_b n$, since $n/b^{\log_b n} = 1$. There are $a^{\log_b n} = n^{\log_b a}$ leaves in the tree.

We can obtain equation (4.21) by summing the costs of the nodes at each depth in the tree, as shown in the figure. The cost for all internal nodes at depth j is $a^{j} f(n/b^{j})$, and so the total cost of all internal nodes is

$$\sum_{j=0}^{\log_b n-1} a^j f(n/b^j) \, .$$

In the underlying divide-and-conquer algorithm, this sum represents the costs of dividing problems into subproblems and then recombining the subproblems. The

cost of all the leaves, which is the cost of doing all $n^{\log_b a}$ subproblems of size 1, is $\Theta(n^{\log_b a})$.

In terms of the recursion tree, the three cases of the master theorem correspond to cases in which the total cost of the tree is (1) dominated by the costs in the leaves, (2) evenly distributed among the levels of the tree, or (3) dominated by the cost of the root.

The summation in equation (4.21) describes the cost of the dividing and combining steps in the underlying divide-and-conquer algorithm. The next lemma provides asymptotic bounds on the summation's growth.

Lemma 4.3

Let $a \ge 1$ and b > 1 be constants, and let f(n) be a nonnegative function defined on exact powers of b. A function g(n) defined over exact powers of b by

$$g(n) = \sum_{j=0}^{\log_b n-1} a^j f(n/b^j)$$
(4.22)

has the following asymptotic bounds for exact powers of b:

- 1. If $f(n) = O(n^{\log_b a \epsilon})$ for some constant $\epsilon > 0$, then $g(n) = O(n^{\log_b a})$.
- 2. If $f(n) = \Theta(n^{\log_b a})$, then $g(n) = \Theta(n^{\log_b a} \lg n)$.
- 3. If $af(n/b) \le cf(n)$ for some constant c < 1 and for all sufficiently large n, then $g(n) = \Theta(f(n))$.

Proof For case 1, we have $f(n) = O(n^{\log_b a - \epsilon})$, which implies that $f(n/b^j) = O((n/b^j)^{\log_b a - \epsilon})$. Substituting into equation (4.22) yields

$$g(n) = O\left(\sum_{j=0}^{\log_b n-1} a^j \left(\frac{n}{b^j}\right)^{\log_b a-\epsilon}\right).$$
(4.23)

We bound the summation within the *O*-notation by factoring out terms and simplifying, which leaves an increasing geometric series:

$$\sum_{j=0}^{\log_b n-1} a^j \left(\frac{n}{b^j}\right)^{\log_b a-\epsilon} = n^{\log_b a-\epsilon} \sum_{j=0}^{\log_b n-1} \left(\frac{ab^\epsilon}{b^{\log_b a}}\right)^j$$
$$= n^{\log_b a-\epsilon} \sum_{j=0}^{\log_b n-1} (b^\epsilon)^j$$
$$= n^{\log_b a-\epsilon} \left(\frac{b^{\epsilon \log_b n} - 1}{b^{\epsilon} - 1}\right)$$

$$= n^{\log_b a - \epsilon} \left(\frac{n^{\epsilon} - 1}{b^{\epsilon} - 1} \right) \,.$$

Since b and ϵ are constants, we can rewrite the last expression as $n^{\log_b a - \epsilon} O(n^{\epsilon}) = O(n^{\log_b a})$. Substituting this expression for the summation in equation (4.23) yields

$$g(n) = O(n^{\log_b a}) ,$$

thereby proving case 1.

Because case 2 assumes that $f(n) = \Theta(n^{\log_b a})$, we have that $f(n/b^j) = \Theta((n/b^j)^{\log_b a})$. Substituting into equation (4.22) yields

$$g(n) = \Theta\left(\sum_{j=0}^{\log_b n-1} a^j \left(\frac{n}{b^j}\right)^{\log_b a}\right) .$$
(4.24)

We bound the summation within the Θ -notation as in case 1, but this time we do not obtain a geometric series. Instead, we discover that every term of the summation is the same:

$$\sum_{j=0}^{\log_b n-1} a^j \left(\frac{n}{b^j}\right)^{\log_b a} = n^{\log_b a} \sum_{j=0}^{\log_b n-1} \left(\frac{a}{b^{\log_b a}}\right)^j$$
$$= n^{\log_b a} \sum_{j=0}^{\log_b n-1} 1$$
$$= n^{\log_b a} \log_b n .$$

Substituting this expression for the summation in equation (4.24) yields

$$g(n) = \Theta(n^{\log_b a} \log_b n) = \Theta(n^{\log_b a} \lg n),$$

proving case 2.

We prove case 3 similarly. Since f(n) appears in the definition (4.22) of g(n)and all terms of g(n) are nonnegative, we can conclude that $g(n) = \Omega(f(n))$ for exact powers of b. We assume in the statement of the lemma that $af(n/b) \le cf(n)$ for some constant c < 1 and all sufficiently large n. We rewrite this assumption as $f(n/b) \le (c/a)f(n)$ and iterate j times, yielding $f(n/b^j) \le (c/a)^j f(n)$ or, equivalently, $a^j f(n/b^j) \le c^j f(n)$, where we assume that the values we iterate on are sufficiently large. Since the last, and smallest, such value is n/b^{j-1} , it is enough to assume that n/b^{j-1} is sufficiently large.

Substituting into equation (4.22) and simplifying yields a geometric series, but unlike the series in case 1, this one has decreasing terms. We use an O(1) term to

capture the terms that are not covered by our assumption that *n* is sufficiently large:

$$g(n) = \sum_{j=0}^{\log_b n-1} a^j f(n/b^j)$$

$$\leq \sum_{j=0}^{\log_b n-1} c^j f(n) + O(1)$$

$$\leq f(n) \sum_{j=0}^{\infty} c^j + O(1)$$

$$= f(n) \left(\frac{1}{1-c}\right) + O(1)$$

$$= O(f(n)),$$

since *c* is a constant. Thus, we can conclude that $g(n) = \Theta(f(n))$ for exact powers of *b*. With case 3 proved, the proof of the lemma is complete.

We can now prove a version of the master theorem for the case in which n is an exact power of b.

Lemma 4.4

Let $a \ge 1$ and b > 1 be constants, and let f(n) be a nonnegative function defined on exact powers of b. Define T(n) on exact powers of b by the recurrence

$$T(n) = \begin{cases} \Theta(1) & \text{if } n = 1, \\ aT(n/b) + f(n) & \text{if } n = b^i \end{cases}$$

where *i* is a positive integer. Then T(n) has the following asymptotic bounds for exact powers of *b*:

- 1. If $f(n) = O(n^{\log_b a \epsilon})$ for some constant $\epsilon > 0$, then $T(n) = \Theta(n^{\log_b a})$.
- 2. If $f(n) = \Theta(n^{\log_b a})$, then $T(n) = \Theta(n^{\log_b a} \lg n)$.
- 3. If $f(n) = \Omega(n^{\log_b a + \epsilon})$ for some constant $\epsilon > 0$, and if $af(n/b) \le cf(n)$ for some constant c < 1 and all sufficiently large *n*, then $T(n) = \Theta(f(n))$.

Proof We use the bounds in Lemma 4.3 to evaluate the summation (4.21) from Lemma 4.2. For case 1, we have

$$T(n) = \Theta(n^{\log_b a}) + O(n^{\log_b a})$$

= $\Theta(n^{\log_b a})$,

and for case 2,

$$T(n) = \Theta(n^{\log_b a}) + \Theta(n^{\log_b a} \lg n)$$

= $\Theta(n^{\log_b a} \lg n)$.

For case 3,

$$T(n) = \Theta(n^{\log_b a}) + \Theta(f(n))$$

= $\Theta(f(n))$,

because $f(n) = \Omega(n^{\log_b a + \epsilon})$.

4.6.2 Floors and ceilings

To complete the proof of the master theorem, we must now extend our analysis to the situation in which floors and ceilings appear in the master recurrence, so that the recurrence is defined for all integers, not for just exact powers of b. Obtaining a lower bound on

$$T(n) = aT(\lceil n/b \rceil) + f(n)$$
(4.25)

and an upper bound on

$$T(n) = aT(\lfloor n/b \rfloor) + f(n)$$
(4.26)

is routine, since we can push through the bound $\lceil n/b \rceil \ge n/b$ in the first case to yield the desired result, and we can push through the bound $\lfloor n/b \rfloor \le n/b$ in the second case. We use much the same technique to lower-bound the recurrence (4.26) as to upper-bound the recurrence (4.25), and so we shall present only this latter bound.

We modify the recursion tree of Figure 4.7 to produce the recursion tree in Figure 4.8. As we go down in the recursion tree, we obtain a sequence of recursive invocations on the arguments

$$[n,b]$$

 $[n/b],b]$
 $[[n/b]/b],$
 $[[[n/b]/b]/b],$
 \vdots

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Let us denote the *j* th element in the sequence by n_j , where

$$n_{j} = \begin{cases} n & \text{if } j = 0, \\ \lceil n_{j-1}/b \rceil & \text{if } j > 0. \end{cases}$$
(4.27)

https://hemanthrajhemu.github.io



Figure 4.8 The recursion tree generated by $T(n) = aT(\lceil n/b \rceil) + f(n)$. The recursive argument n_j is given by equation (4.27).

Our first goal is to determine the depth k such that n_k is a constant. Using the inequality $\lceil x \rceil \le x + 1$, we obtain

$$n_{0} \leq n,$$

$$n_{1} \leq \frac{n}{b} + 1,$$

$$n_{2} \leq \frac{n}{b^{2}} + \frac{1}{b} + 1,$$

$$n_{3} \leq \frac{n}{b^{3}} + \frac{1}{b^{2}} + \frac{1}{b} + 1,$$

$$\vdots$$

In general, we have

$$n_j \leq \frac{n}{b^j} + \sum_{i=0}^{j-1} \frac{1}{b^i}$$
$$< \frac{n}{b^j} + \sum_{i=0}^{\infty} \frac{1}{b^i}$$
$$= \frac{n}{b^j} + \frac{b}{b-1}.$$

Letting $j = \lfloor \log_b n \rfloor$, we obtain

$$n_{\lfloor \log_b n \rfloor} < \frac{n}{b^{\lfloor \log_b n \rfloor}} + \frac{b}{b-1}$$
$$< \frac{n}{b^{\log_b n-1}} + \frac{b}{b-1}$$
$$= \frac{n}{n/b} + \frac{b}{b-1}$$
$$= b + \frac{b}{b-1}$$
$$= O(1),$$

and thus we see that at depth $\lfloor \log_b n \rfloor$, the problem size is at most a constant.

From Figure 4.8, we see that

$$T(n) = \Theta(n^{\log_b a}) + \sum_{j=0}^{\lfloor \log_b a \rfloor - 1} a^j f(n_j) , \qquad (4.28)$$

which is much the same as equation (4.21), except that n is an arbitrary integer and not restricted to be an exact power of b.

We can now evaluate the summation

$$g(n) = \sum_{j=0}^{\lfloor \log_b n \rfloor - 1} a^j f(n_j)$$
(4.29)

from equation (4.28) in a manner analogous to the proof of Lemma 4.3. Beginning with case 3, if $af(\lceil n/b \rceil) \leq cf(n)$ for n > b+b/(b-1), where c < 1 is a constant, then it follows that $a^j f(n_j) \leq c^j f(n)$. Therefore, we can evaluate the sum in equation (4.29) just as in Lemma 4.3. For case 2, we have $f(n) = \Theta(n^{\log_b a})$. If we can show that $f(n_j) = O(n^{\log_b a}/a^j) = O((n/b^j)^{\log_b a})$, then the proof for case 2 of Lemma 4.3 will go through. Observe that $j \leq \lfloor \log_b n \rfloor$ implies $b^j/n \leq 1$. The bound $f(n) = O(n^{\log_b a})$ implies that there exists a constant c > 0 such that for all sufficiently large n_j ,

$$\begin{split} f(n_j) &\leq c \left(\frac{n}{b^j} + \frac{b}{b-1}\right)^{\log_b a} \\ &= c \left(\frac{n}{b^j} \left(1 + \frac{b^j}{n} \cdot \frac{b}{b-1}\right)\right)^{\log_b a} \\ &= c \left(\frac{n^{\log_b a}}{a^j}\right) \left(1 + \left(\frac{b^j}{n} \cdot \frac{b}{b-1}\right)\right)^{\log_b a} \\ &\leq c \left(\frac{n^{\log_b a}}{a^j}\right) \left(1 + \frac{b}{b-1}\right)^{\log_b a} \\ &= O\left(\frac{n^{\log_b a}}{a^j}\right), \end{split}$$

since $c(1 + b/(b-1))^{\log_b a}$ is a constant. Thus, we have proved case 2. The proof of case 1 is almost identical. The key is to prove the bound $f(n_j) = O(n^{\log_b a-\epsilon})$, which is similar to the corresponding proof of case 2, though the algebra is more intricate.

We have now proved the upper bounds in the master theorem for all integers n. The proof of the lower bounds is similar.

Exercises

4.6-1 *

Give a simple and exact expression for n_j in equation (4.27) for the case in which *b* is a positive integer instead of an arbitrary real number.

4.6-2 *****

Show that if $f(n) = \Theta(n^{\log_b a} \lg^k n)$, where $k \ge 0$, then the master recurrence has solution $T(n) = \Theta(n^{\log_b a} \lg^{k+1} n)$. For simplicity, confine your analysis to exact powers of *b*.

4.6-3 *

Show that case 3 of the master theorem is overstated, in the sense that the regularity condition $af(n/b) \le cf(n)$ for some constant c < 1 implies that there exists a constant $\epsilon > 0$ such that $f(n) = \Omega(n^{\log_b a + \epsilon})$.

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Problems

4-1 Recurrence examples

Give asymptotic upper and lower bounds for T(n) in each of the following recurrences. Assume that T(n) is constant for $n \leq 2$. Make your bounds as tight as possible, and justify your answers.

a. $T(n) = 2T(n/2) + n^4$.

b.
$$T(n) = T(7n/10) + n$$
.

- c. $T(n) = 16T(n/4) + n^2$.
- *d.* $T(n) = 7T(n/3) + n^2$.
- e. $T(n) = 7T(n/2) + n^2$.
- f. $T(n) = 2T(n/4) + \sqrt{n}$.

g.
$$T(n) = T(n-2) + n^2$$
.

4-2 Parameter-passing costs

Throughout this book, we assume that parameter passing during procedure calls takes constant time, even if an N-element array is being passed. This assumption is valid in most systems because a pointer to the array is passed, not the array itself. This problem examines the implications of three parameter-passing strategies:

- 1. An array is passed by pointer. Time = $\Theta(1)$.
- 2. An array is passed by copying. Time = $\Theta(N)$, where N is the size of the array.
- 3. An array is passed by copying only the subrange that might be accessed by the called procedure. Time = $\Theta(q p + 1)$ if the subarray $A[p \dots q]$ is passed.
- *a.* Consider the recursive binary search algorithm for finding a number in a sorted array (see Exercise 2.3-5). Give recurrences for the worst-case running times of binary search when arrays are passed using each of the three methods above, and give good upper bounds on the solutions of the recurrences. Let N be the size of the original problem and n be the size of a subproblem.
- **b.** Redo part (a) for the MERGE-SORT algorithm from Section 2.3.1.

4-3 More recurrence examples

Give asymptotic upper and lower bounds for T(n) in each of the following recurrences. Assume that T(n) is constant for sufficiently small n. Make your bounds as tight as possible, and justify your answers.

a.
$$T(n) = 4T(n/3) + n \lg n$$
.

b.
$$T(n) = 3T(n/3) + n/\lg n$$
.

c.
$$T(n) = 4T(n/2) + n^2\sqrt{n}$$
.

d.
$$T(n) = 3T(n/3 - 2) + n/2$$
.

- e. $T(n) = 2T(n/2) + n/\lg n$.
- f. T(n) = T(n/2) + T(n/4) + T(n/8) + n.
- g. T(n) = T(n-1) + 1/n.
- **h.** $T(n) = T(n-1) + \lg n$.

i.
$$T(n) = T(n-2) + 1/\lg n$$
.

$$j. \quad T(n) = \sqrt{n}T(\sqrt{n}) + n.$$

4-4 Fibonacci numbers

This problem develops properties of the Fibonacci numbers, which are defined by recurrence (3.22). We shall use the technique of generating functions to solve the Fibonacci recurrence. Define the *generating function* (or *formal power series*) \mathcal{F} as

$$\begin{aligned} \mathcal{F}(z) &= \sum_{i=0}^{\infty} F_i z^i \\ &= 0 + z + z^2 + 2z^3 + 3z^4 + 5z^5 + 8z^6 + 13z^7 + 21z^8 + \cdots, \end{aligned}$$

where F_i is the *i*th Fibonacci number.

a. Show that $\mathcal{F}(z) = z + z \mathcal{F}(z) + z^2 \mathcal{F}(z)$.

b. Show that

$$\begin{aligned} \mathcal{F}(z) &= \frac{z}{1 - z - z^2} \\ &= \frac{z}{(1 - \phi z)(1 - \hat{\phi} z)} \\ &= \frac{1}{\sqrt{5}} \left(\frac{1}{1 - \phi z} - \frac{1}{1 - \hat{\phi} z} \right) \,, \end{aligned}$$

where

$$\phi = \frac{1 + \sqrt{5}}{2} = 1.61803\dots$$

and

$$\hat{\phi} = \frac{1 - \sqrt{5}}{2} = -0.61803\dots$$

c. Show that

$$\mathcal{F}(z) = \sum_{i=0}^{\infty} \frac{1}{\sqrt{5}} (\phi^i - \hat{\phi}^i) z^i$$

d. Use part (c) to prove that $F_i = \phi^i / \sqrt{5}$ for i > 0, rounded to the nearest integer. (*Hint:* Observe that $|\hat{\phi}| < 1$.)

4-5 Chip testing

Professor Diogenes has n supposedly identical integrated-circuit chips that in principle are capable of testing each other. The professor's test jig accommodates two chips at a time. When the jig is loaded, each chip tests the other and reports whether it is good or bad. A good chip always reports accurately whether the other chip is good or bad, but the professor cannot trust the answer of a bad chip. Thus, the four possible outcomes of a test are as follows:

Chip A says	Chip B says	Conclusion
B is good	A is good	both are good, or both are bad
B is good	A is bad	at least one is bad
B is bad	A is good	at least one is bad
B is bad	A is bad	at least one is bad

a. Show that if more than n/2 chips are bad, the professor cannot necessarily determine which chips are good using any strategy based on this kind of pairwise test. Assume that the bad chips can conspire to fool the professor.

- **b.** Consider the problem of finding a single good chip from among *n* chips, assuming that more than n/2 of the chips are good. Show that $\lfloor n/2 \rfloor$ pairwise tests are sufficient to reduce the problem to one of nearly half the size.
- c. Show that the good chips can be identified with $\Theta(n)$ pairwise tests, assuming that more than n/2 of the chips are good. Give and solve the recurrence that describes the number of tests.

4-6 Monge arrays

An $m \times n$ array A of real numbers is a **Monge array** if for all i, j, k, and l such that $1 \le i < k \le m$ and $1 \le j < l \le n$, we have

 $A[i, j] + A[k, l] \le A[i, l] + A[k, j].$

In other words, whenever we pick two rows and two columns of a Monge array and consider the four elements at the intersections of the rows and the columns, the sum of the upper-left and lower-right elements is less than or equal to the sum of the lower-left and upper-right elements. For example, the following array is Monge:

10	17	13	28	23
17	22	16	29	23
24	28	22	34	24
11	13	6	17	7
45	44	32	37	23
36	33	19	21	6
75	66	51	53	34

a. Prove that an array is Monge if and only if for all i = 1, 2, ..., m - 1 and j = 1, 2, ..., m - 1, we have

 $A[i, j] + A[i + 1, j + 1] \le A[i, j + 1] + A[i + 1, j].$

(*Hint:* For the "if" part, use induction separately on rows and columns.)

- *b.* The following array is not Monge. Change one element in order to make it Monge. (*Hint:* Use part (a).)

- c. Let f(i) be the index of the column containing the leftmost minimum element of row *i*. Prove that $f(1) \le f(2) \le \cdots \le f(m)$ for any $m \times n$ Monge array.
- *d.* Here is a description of a divide-and-conquer algorithm that computes the leftmost minimum element in each row of an $m \times n$ Monge array A:

Construct a submatrix A' of A consisting of the even-numbered rows of A. Recursively determine the leftmost minimum for each row of A'. Then compute the leftmost minimum in the odd-numbered rows of A.

Explain how to compute the leftmost minimum in the odd-numbered rows of A (given that the leftmost minimum of the even-numbered rows is known) in O(m + n) time.

e. Write the recurrence describing the running time of the algorithm described in part (d). Show that its solution is $O(m + n \log m)$.

Chapter notes

Divide-and-conquer as a technique for designing algorithms dates back to at least 1962 in an article by Karatsuba and Ofman [194]. It might have been used well before then, however; according to Heideman, Johnson, and Burrus [163], C. F. Gauss devised the first fast Fourier transform algorithm in 1805, and Gauss's formulation breaks the problem into smaller subproblems whose solutions are combined.

The maximum-subarray problem in Section 4.1 is a minor variation on a problem studied by Bentley [43, Chapter 7].

Strassen's algorithm [325] caused much excitement when it was published in 1969. Before then, few imagined the possibility of an algorithm asymptotically faster than the basic SQUARE-MATRIX-MULTIPLY procedure. The asymptotic upper bound for matrix multiplication has been improved since then. The most asymptotically efficient algorithm for multiplying $n \times n$ matrices to date, due to Coppersmith and Winograd [78], has a running time of $O(n^{2.376})$. The best lower bound known is just the obvious $\Omega(n^2)$ bound (obvious because we must fill in n^2 elements of the product matrix).

From a practical point of view, Strassen's algorithm is often not the method of choice for matrix multiplication, for four reasons:

- 1. The constant factor hidden in the $\Theta(n^{\lg 7})$ running time of Strassen's algorithm is larger than the constant factor in the $\Theta(n^3)$ -time SQUARE-MATRIX-MULTIPLY procedure.
- 2. When the matrices are sparse, methods tailored for sparse matrices are faster.



- 3. Strassen's algorithm is not quite as numerically stable as SQUARE-MATRIX-MULTIPLY. In other words, because of the limited precision of computer arithmetic on noninteger values, larger errors accumulate in Strassen's algorithm than in SQUARE-MATRIX-MULTIPLY.
- 4. The submatrices formed at the levels of recursion consume space.

The latter two reasons were mitigated around 1990. Higham [167] demonstrated that the difference in numerical stability had been overemphasized; although Strassen's algorithm is too numerically unstable for some applications, it is within acceptable limits for others. Bailey, Lee, and Simon [32] discuss techniques for reducing the memory requirements for Strassen's algorithm.

In practice, fast matrix-multiplication implementations for dense matrices use Strassen's algorithm for matrix sizes above a "crossover point," and they switch to a simpler method once the subproblem size reduces to below the crossover point. The exact value of the crossover point is highly system dependent. Analyses that count operations but ignore effects from caches and pipelining have produced crossover points as low as n = 8 (by Higham [167]) or n = 12 (by Huss-Lederman et al. [186]). D'Alberto and Nicolau [81] developed an adaptive scheme, which determines the crossover point by benchmarking when their software package is installed. They found crossover points on various systems ranging from n = 400to n = 2150, and they could not find a crossover point on a couple of systems.

Recurrences were studied as early as 1202 by L. Fibonacci, for whom the Fibonacci numbers are named. A. De Moivre introduced the method of generating functions (see Problem 4-4) for solving recurrences. The master method is adapted from Bentley, Haken, and Saxe [44], which provides the extended method justified by Exercise 4.6-2. Knuth [209] and Liu [237] show how to solve linear recurrences using the method of generating functions. Purdom and Brown [287] and Graham, Knuth, and Patashnik [152] contain extended discussions of recurrence solving.

Several researchers, including Akra and Bazzi [13], Roura [299], Verma [346], and Yap [360], have given methods for solving more general divide-and-conquer recurrences than are solved by the master method. We describe the result of Akra and Bazzi here, as modified by Leighton [228]. The Akra-Bazzi method works for recurrences of the form

$$T(x) = \begin{cases} \Theta(1) & \text{if } 1 \le x \le x_0 ,\\ \sum_{i=1}^k a_i T(b_i x) + f(x) & \text{if } x > x_0 , \end{cases}$$
(4.30)

where

- $x \ge 1$ is a real number,
- x_0 is a constant such that $x_0 \ge 1/b_i$ and $x_0 \ge 1/(1-b_i)$ for i = 1, 2, ..., k,
- a_i is a positive constant for i = 1, 2, ..., k,

- b_i is a constant in the range $0 < b_i < 1$ for i = 1, 2, ..., k,
- $k \ge 1$ is an integer constant, and
- f(x) is a nonnegative function that satisfies the *polynomial-growth condi*tion: there exist positive constants c_1 and c_2 such that for all $x \ge 1$, for i = 1, 2, ..., k, and for all u such that $b_i x \le u \le x$, we have $c_1 f(x) \le f(u) \le c_2 f(x)$. (If |f'(x)| is upper-bounded by some polynomial in x, then f(x) satisfies the polynomial-growth condition. For example, $f(x) = x^{\alpha} \lg^{\beta} x$ satisfies this condition for any real constants α and β .)

Although the master method does not apply to a recurrence such as $T(n) = T(\lfloor n/3 \rfloor) + T(\lfloor 2n/3 \rfloor) + O(n)$, the Akra-Bazzi method does. To solve the recurrence (4.30), we first find the unique real number p such that $\sum_{i=1}^{k} a_i b_i^p = 1$. (Such a p always exists.) The solution to the recurrence is then

$$T(n) = \Theta\left(x^p\left(1 + \int_1^x \frac{f(u)}{u^{p+1}} du\right)\right)$$

The Akra-Bazzi method can be somewhat difficult to use, but it serves in solving recurrences that model division of the problem into substantially unequally sized subproblems. The master method is simpler to use, but it applies only when sub-problem sizes are equal.

17 Amortized Analysis

In an *amortized analysis*, we average the time required to perform a sequence of data-structure operations over all the operations performed. With amortized analysis, we can show that the average cost of an operation is small, if we average over a sequence of operations, even though a single operation within the sequence might be expensive. Amortized analysis differs from average-case analysis in that probability is not involved; an amortized analysis guarantees the *average performance of each operation in the worst case*.

The first three sections of this chapter cover the three most common techniques used in amortized analysis. Section 17.1 starts with aggregate analysis, in which we determine an upper bound T(n) on the total cost of a sequence of n operations. The average cost per operation is then T(n)/n. We take the average cost as the amortized cost of each operation, so that all operations have the same amortized cost.

Section 17.2 covers the accounting method, in which we determine an amortized cost of each operation. When there is more than one type of operation, each type of operation may have a different amortized cost. The accounting method overcharges some operations early in the sequence, storing the overcharge as "prepaid credit" on specific objects in the data structure. Later in the sequence, the credit pays for operations that are charged less than they actually cost.

Section 17.3 discusses the potential method, which is like the accounting method in that we determine the amortized cost of each operation and may overcharge operations early on to compensate for undercharges later. The potential method maintains the credit as the "potential energy" of the data structure as a whole instead of associating the credit with individual objects within the data structure.

We shall use two examples to examine these three methods. One is a stack with the additional operation MULTIPOP, which pops several objects at once. The other is a binary counter that counts up from 0 by means of the single operation INCREMENT.

While reading this chapter, bear in mind that the charges assigned during an amortized analysis are for analysis purposes only. They need not—and should not—appear in the code. If, for example, we assign a credit to an object x when using the accounting method, we have no need to assign an appropriate amount to some attribute, such as x. *credit*, in the code.

When we perform an amortized analysis, we often gain insight into a particular data structure, and this insight can help us optimize the design. In Section 17.4, for example, we shall use the potential method to analyze a dynamically expanding and contracting table.

17.1 Aggregate analysis

In *aggregate analysis*, we show that for all n, a sequence of n operations takes *worst-case* time T(n) in total. In the worst case, the average cost, or *amortized cost*, per operation is therefore T(n)/n. Note that this amortized cost applies to each operation, even when there are several types of operations in the sequence. The other two methods we shall study in this chapter, the accounting method and the potential method, may assign different amortized costs to different types of operations.

Stack operations

In our first example of aggregate analysis, we analyze stacks that have been augmented with a new operation. Section 10.1 presented the two fundamental stack operations, each of which takes O(1) time:

PUSH(S, x) pushes object x onto stack S.

POP(S) pops the top of stack S and returns the popped object. Calling POP on an empty stack generates an error.

Since each of these operations runs in O(1) time, let us consider the cost of each to be 1. The total cost of a sequence of *n* PUSH and POP operations is therefore *n*, and the actual running time for *n* operations is therefore $\Theta(n)$.

Now we add the stack operation MULTIPOP(S, k), which removes the k top objects of stack S, popping the entire stack if the stack contains fewer than k objects. Of course, we assume that k is positive; otherwise the MULTIPOP operation leaves the stack unchanged. In the following pseudocode, the operation STACK-EMPTY returns TRUE if there are no objects currently on the stack, and FALSE otherwise.

top → 23		
17		
6		
39		
10	top → 10	
47	47	
(a)	(b)	(c)

Figure 17.1 The action of MULTIPOP on a stack *S*, shown initially in (a). The top 4 objects are popped by MULTIPOP(*S*, 4), whose result is shown in (b). The next operation is MULTIPOP(*S*, 7), which empties the stack—shown in (c)—since there were fewer than 7 objects remaining.

MULTIPOP(S,k)

- 1 while not STACK-EMPTY(S) and k > 0
- 2 POP(S)
- 3 k = k 1

Figure 17.1 shows an example of MULTIPOP.

What is the running time of MULTIPOP(S, k) on a stack of s objects? The actual running time is linear in the number of POP operations actually executed, and thus we can analyze MULTIPOP in terms of the abstract costs of 1 each for PUSH and POP. The number of iterations of the **while** loop is the number min(s, k) of objects popped off the stack. Each iteration of the loop makes one call to POP in line 2. Thus, the total cost of MULTIPOP is min(s, k), and the actual running time is a linear function of this cost.

Let us analyze a sequence of *n* PUSH, POP, and MULTIPOP operations on an initially empty stack. The worst-case cost of a MULTIPOP operation in the sequence is O(n), since the stack size is at most *n*. The worst-case time of any stack operation is therefore O(n), and hence a sequence of *n* operations costs $O(n^2)$, since we may have O(n) MULTIPOP operations costing O(n) each. Although this analysis is correct, the $O(n^2)$ result, which we obtained by considering the worst-case cost of each operation individually, is not tight.

Using aggregate analysis, we can obtain a better upper bound that considers the entire sequence of n operations. In fact, although a single MULTIPOP operation can be expensive, any sequence of n PUSH, POP, and MULTIPOP operations on an initially empty stack can cost at most O(n). Why? We can pop each object from the stack at most once for each time we have pushed it onto the stack. Therefore, the number of times that POP can be called on a nonempty stack, including calls within MULTIPOP, is at most the number of PUSH operations, which is at most n. For any value of n, any sequence of n PUSH, POP, and MULTIPOP operations takes a total of O(n) time. The average cost of an operation is O(n)/n = O(1). In aggregate



analysis, we assign the amortized cost of each operation to be the average cost. In this example, therefore, all three stack operations have an amortized cost of O(1).

We emphasize again that although we have just shown that the average cost, and hence the running time, of a stack operation is O(1), we did not use probabilistic reasoning. We actually showed a *worst-case* bound of O(n) on a sequence of n operations. Dividing this total cost by n yielded the average cost per operation, or the amortized cost.

Incrementing a binary counter

As another example of aggregate analysis, consider the problem of implementing a *k*-bit binary counter that counts upward from 0. We use an array A[0..k-1] of bits, where A.length = k, as the counter. A binary number *x* that is stored in the counter has its lowest-order bit in A[0] and its highest-order bit in A[k-1], so that $x = \sum_{i=0}^{k-1} A[i] \cdot 2^i$. Initially, x = 0, and thus A[i] = 0 for i = 0, 1, ..., k-1. To add 1 (modulo 2^k) to the value in the counter, we use the following procedure.

INCREMENT(A)

 $\begin{array}{ll}
1 & i = 0 \\
2 & \text{while } i < A. length \text{ and } A[i] == 1 \\
3 & A[i] = 0 \\
4 & i = i + 1 \\
5 & \text{if } i < A. length \\
6 & A[i] = 1
\end{array}$

Figure 17.2 shows what happens to a binary counter as we increment it 16 times, starting with the initial value 0 and ending with the value 16. At the start of each iteration of the **while** loop in lines 2–4, we wish to add a 1 into position i. If A[i] = 1, then adding 1 flips the bit to 0 in position i and yields a carry of 1, to be added into position i + 1 on the next iteration of the loop. Otherwise, the loop ends, and then, if i < k, we know that A[i] = 0, so that line 6 adds a 1 into position i, flipping the 0 to a 1. The cost of each INCREMENT operation is linear in the number of bits flipped.

As with the stack example, a cursory analysis yields a bound that is correct but not tight. A single execution of INCREMENT takes time $\Theta(k)$ in the worst case, in which array A contains all 1s. Thus, a sequence of n INCREMENT operations on an initially zero counter takes time O(nk) in the worst case.

We can tighten our analysis to yield a worst-case cost of O(n) for a sequence of nINCREMENT operations by observing that not all bits flip each time INCREMENT is called. As Figure 17.2 shows, A[0] does flip each time INCREMENT is called. The next bit up, A[1], flips only every other time: a sequence of n INCREMENT

Counter value	AT AT AT AT ALL ALL ALL ALL ALL ALL ALL	Total cost
0	0 0 0 0 0 0 0 0	0
1	0 0 0 0 0 0 0 1	1
2	0 0 0 0 0 0 1 0	3
3	0 0 0 0 0 0 1 1	4
4	0 0 0 0 0 1 0 0	7
5	0 0 0 0 0 1 0 1	8
6	0 0 0 0 0 1 1 0	10
7	0 0 0 0 0 1 1 1	11
8	0 0 0 0 1 0 0 0	15
9	0 0 0 0 1 0 0 1	16
10	0 0 0 0 1 0 1 0	18
11	0 0 0 0 1 0 1 1	19
12	0 0 0 0 1 1 0 0	22
13	0 0 0 0 1 1 0 1	23
14	0 0 0 0 1 1 1 0	25
15	0 0 0 0 1 1 1 1	26
16	0 0 0 1 0 0 0 0	31

Figure 17.2 An 8-bit binary counter as its value goes from 0 to 16 by a sequence of 16 INCREMENT operations. Bits that flip to achieve the next value are shaded. The running cost for flipping bits is shown at the right. Notice that the total cost is always less than twice the total number of INCREMENT operations.

operations on an initially zero counter causes A[1] to flip $\lfloor n/2 \rfloor$ times. Similarly, bit A[2] flips only every fourth time, or $\lfloor n/4 \rfloor$ times in a sequence of *n* INCREMENT operations. In general, for i = 0, 1, ..., k - 1, bit A[i] flips $\lfloor n/2^i \rfloor$ times in a sequence of *n* INCREMENT operations on an initially zero counter. For $i \ge k$, bit A[i] does not exist, and so it cannot flip. The total number of flips in the sequence is thus

$$\sum_{i=0}^{k-1} \left\lfloor \frac{n}{2^i} \right\rfloor < n \sum_{i=0}^{\infty} \frac{1}{2^i}$$
$$= 2n ,$$

by equation (A.6). The worst-case time for a sequence of *n* INCREMENT operations on an initially zero counter is therefore O(n). The average cost of each operation, and therefore the amortized cost per operation, is O(n)/n = O(1).
Exercises

17.1-1

If the set of stack operations included a MULTIPUSH operation, which pushes k items onto the stack, would the O(1) bound on the amortized cost of stack operations continue to hold?

17.1-2

Show that if a DECREMENT operation were included in the *k*-bit counter example, n operations could cost as much as $\Theta(nk)$ time.

17.1-3

Suppose we perform a sequence of n operations on a data structure in which the i th operation costs i if i is an exact power of 2, and 1 otherwise. Use aggregate analysis to determine the amortized cost per operation.

17.2 The accounting method

In the *accounting method* of amortized analysis, we assign differing charges to different operations, with some operations charged more or less than they actually cost. We call the amount we charge an operation its *amortized cost*. When an operation's amortized cost exceeds its actual cost, we assign the difference to specific objects in the data structure as *credit*. Credit can help pay for later operations whose amortized cost is less than their actual cost. Thus, we can view the amortized cost of an operation as being split between its actual cost and credit that is either deposited or used up. Different operations may have different amortized costs. This method differs from aggregate analysis, in which all operations have the same amortized cost.

We must choose the amortized costs of operations carefully. If we want to show that in the worst case the average cost per operation is small by analyzing with amortized costs, we must ensure that the total amortized cost of a sequence of operations provides an upper bound on the total actual cost of the sequence. Moreover, as in aggregate analysis, this relationship must hold for all sequences of operations. If we denote the actual cost of the *i*th operation by c_i and the amortized cost of the *i*th operation by \hat{c}_i , we require

$$\sum_{i=1}^{n} \hat{c}_i \ge \sum_{i=1}^{n} c_i \tag{17.1}$$

for all sequences of n operations. The total credit stored in the data structure is the difference between the total amortized cost and the total actual cost, or

 $\sum_{i=1}^{n} \hat{c}_i - \sum_{i=1}^{n} c_i$. By inequality (17.1), the total credit associated with the data structure must be nonnegative at all times. If we ever were to allow the total credit to become negative (the result of undercharging early operations with the promise of repaying the account later on), then the total amortized costs incurred at that time would be below the total actual costs incurred; for the sequence of operations up to that time, the total amortized cost would not be an upper bound on the total actual cost. Thus, we must take care that the total credit in the data structure never becomes negative.

Stack operations

To illustrate the accounting method of amortized analysis, let us return to the stack example. Recall that the actual costs of the operations were

```
PUSH1POP1MULTIPOP\min(k, s)
```

where k is the argument supplied to MULTIPOP and s is the stack size when it is called. Let us assign the following amortized costs:

```
PUSH2 ,POP0 ,MULTIPOP0 .
```

Note that the amortized cost of MULTIPOP is a constant (0), whereas the actual cost is variable. Here, all three amortized costs are constant. In general, the amortized costs of the operations under consideration may differ from each other, and they may even differ asymptotically.

We shall now show that we can pay for any sequence of stack operations by charging the amortized costs. Suppose we use a dollar bill to represent each unit of cost. We start with an empty stack. Recall the analogy of Section 10.1 between the stack data structure and a stack of plates in a cafeteria. When we push a plate on the stack, we use 1 dollar to pay the actual cost of the push and are left with a credit of 1 dollar (out of the 2 dollars charged), which we leave on top of the plate. At any point in time, every plate on the stack has a dollar of credit on it.

The dollar stored on the plate serves as prepayment for the cost of popping it from the stack. When we execute a POP operation, we charge the operation nothing and pay its actual cost using the credit stored in the stack. To pop a plate, we take the dollar of credit off the plate and use it to pay the actual cost of the operation. Thus, by charging the PUSH operation a little bit more, we can charge the POP operation nothing.

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Moreover, we can also charge MULTIPOP operations nothing. To pop the first plate, we take the dollar of credit off the plate and use it to pay the actual cost of a POP operation. To pop a second plate, we again have a dollar of credit on the plate to pay for the POP operation, and so on. Thus, we have always charged enough up front to pay for MULTIPOP operations. In other words, since each plate on the stack has 1 dollar of credit on it, and the stack always has a nonnegative number of plates, we have ensured that the amount of credit is always nonnegative. Thus, for *any* sequence of *n* PUSH, POP, and MULTIPOP operations, the total amortized cost is O(n), so is the total actual cost.

Incrementing a binary counter

As another illustration of the accounting method, we analyze the INCREMENT operation on a binary counter that starts at zero. As we observed earlier, the running time of this operation is proportional to the number of bits flipped, which we shall use as our cost for this example. Let us once again use a dollar bill to represent each unit of cost (the flipping of a bit in this example).

For the amortized analysis, let us charge an amortized cost of 2 dollars to set a bit to 1. When a bit is set, we use 1 dollar (out of the 2 dollars charged) to pay for the actual setting of the bit, and we place the other dollar on the bit as credit to be used later when we flip the bit back to 0. At any point in time, every 1 in the counter has a dollar of credit on it, and thus we can charge nothing to reset a bit to 0; we just pay for the reset with the dollar bill on the bit.

Now we can determine the amortized cost of INCREMENT. The cost of resetting the bits within the **while** loop is paid for by the dollars on the bits that are reset. The INCREMENT procedure sets at most one bit, in line 6, and therefore the amortized cost of an INCREMENT operation is at most 2 dollars. The number of 1s in the counter never becomes negative, and thus the amount of credit stays nonnegative at all times. Thus, for *n* INCREMENT operations, the total amortized cost is O(n), which bounds the total actual cost.

Exercises

17.2-1

Suppose we perform a sequence of stack operations on a stack whose size never exceeds k. After every k operations, we make a copy of the entire stack for backup purposes. Show that the cost of n stack operations, including copying the stack, is O(n) by assigning suitable amortized costs to the various stack operations.

17.2-2

Redo Exercise 17.1-3 using an accounting method of analysis.

17.2-3

Suppose we wish not only to increment a counter but also to reset it to zero (i.e., make all bits in it 0). Counting the time to examine or modify a bit as $\Theta(1)$, show how to implement a counter as an array of bits so that any sequence of *n* INCREMENT and RESET operations takes time O(n) on an initially zero counter. (*Hint:* Keep a pointer to the high-order 1.)

17.3 The potential method

Instead of representing prepaid work as credit stored with specific objects in the data structure, the *potential method* of amortized analysis represents the prepaid work as "potential energy," or just "potential," which can be released to pay for future operations. We associate the potential with the data structure as a whole rather than with specific objects within the data structure.

The potential method works as follows. We will perform *n* operations, starting with an initial data structure D_0 . For each i = 1, 2, ..., n, we let c_i be the actual cost of the *i*th operation and D_i be the data structure that results after applying the *i*th operation to data structure D_{i-1} . A *potential function* Φ maps each data structure D_i to a real number $\Phi(D_i)$, which is the *potential* associated with data structure D_i . The *amortized cost* \hat{c}_i of the *i*th operation with respect to potential function Φ is defined by

$$\hat{c}_i = c_i + \Phi(D_i) - \Phi(D_{i-1}).$$
(17.2)

The amortized cost of each operation is therefore its actual cost plus the change in potential due to the operation. By equation (17.2), the total amortized cost of the *n* operations is

$$\sum_{i=1}^{n} \hat{c}_{i} = \sum_{i=1}^{n} (c_{i} + \Phi(D_{i}) - \Phi(D_{i-1}))$$
$$= \sum_{i=1}^{n} c_{i} + \Phi(D_{n}) - \Phi(D_{0}).$$
(17.3)

The second equality follows from equation (A.9) because the $\Phi(D_i)$ terms telescope.

If we can define a potential function Φ so that $\Phi(D_n) \ge \Phi(D_0)$, then the total amortized cost $\sum_{i=1}^{n} \hat{c}_i$ gives an upper bound on the total actual cost $\sum_{i=1}^{n} c_i$.

In practice, we do not always know how many operations might be performed. Therefore, if we require that $\Phi(D_i) \ge \Phi(D_0)$ for all *i*, then we guarantee, as in the accounting method, that we pay in advance. We usually just define $\Phi(D_0)$ to be 0 and then show that $\Phi(D_i) \ge 0$ for all *i*. (See Exercise 17.3-1 for an easy way to handle cases in which $\Phi(D_0) \ne 0$.)

Intuitively, if the potential difference $\Phi(D_i) - \Phi(D_{i-1})$ of the *i*th operation is positive, then the amortized cost \hat{c}_i represents an overcharge to the *i*th operation, and the potential of the data structure increases. If the potential difference is negative, then the amortized cost represents an undercharge to the *i*th operation, and the decrease in the potential pays for the actual cost of the operation.

The amortized costs defined by equations (17.2) and (17.3) depend on the choice of the potential function Φ . Different potential functions may yield different amortized costs yet still be upper bounds on the actual costs. We often find trade-offs that we can make in choosing a potential function; the best potential function to use depends on the desired time bounds.

Stack operations

To illustrate the potential method, we return once again to the example of the stack operations PUSH, POP, and MULTIPOP. We define the potential function Φ on a stack to be the number of objects in the stack. For the empty stack D_0 with which we start, we have $\Phi(D_0) = 0$. Since the number of objects in the stack is never negative, the stack D_i that results after the *i* th operation has nonnegative potential, and thus

$$\Phi(D_i) \geq 0 \\ = \Phi(D_0)$$

The total amortized cost of *n* operations with respect to Φ therefore represents an upper bound on the actual cost.

Let us now compute the amortized costs of the various stack operations. If the i th operation on a stack containing s objects is a PUSH operation, then the potential difference is

$$\Phi(D_i) - \Phi(D_{i-1}) = (s+1) - s$$

= 1.

By equation (17.2), the amortized cost of this PUSH operation is

$$\hat{c}_i = c_i + \Phi(D_i) - \Phi(D_{i-1})$$

= 1 + 1
= 2.

Suppose that the *i*th operation on the stack is MULTIPOP(S, k), which causes $k' = \min(k, s)$ objects to be popped off the stack. The actual cost of the operation is k', and the potential difference is

$$\Phi(D_i) - \Phi(D_{i-1}) = -k'.$$

Thus, the amortized cost of the MULTIPOP operation is

$$\hat{c}_i = c_i + \Phi(D_i) - \Phi(D_{i-1})$$

= $k' - k'$
= 0.

Similarly, the amortized cost of an ordinary POP operation is 0.

The amortized cost of each of the three operations is O(1), and thus the total amortized cost of a sequence of *n* operations is O(n). Since we have already argued that $\Phi(D_i) \ge \Phi(D_0)$, the total amortized cost of *n* operations is an upper bound on the total actual cost. The worst-case cost of *n* operations is therefore O(n).

Incrementing a binary counter

As another example of the potential method, we again look at incrementing a binary counter. This time, we define the potential of the counter after the *i* th INCREMENT operation to be b_i , the number of 1s in the counter after the *i* th operation.

Let us compute the amortized cost of an INCREMENT operation. Suppose that the *i*th INCREMENT operation resets t_i bits. The actual cost of the operation is therefore at most $t_i + 1$, since in addition to resetting t_i bits, it sets at most one bit to 1. If $b_i = 0$, then the *i*th operation resets all *k* bits, and so $b_{i-1} = t_i = k$. If $b_i > 0$, then $b_i = b_{i-1} - t_i + 1$. In either case, $b_i \le b_{i-1} - t_i + 1$, and the potential difference is

$$\Phi(D_i) - \Phi(D_{i-1}) \leq (b_{i-1} - t_i + 1) - b_{i-1}$$

= 1 - t_i.

The amortized cost is therefore

$$\hat{c}_i = c_i + \Phi(D_i) - \Phi(D_{i-1}) \\
\leq (t_i + 1) + (1 - t_i) \\
= 2.$$

If the counter starts at zero, then $\Phi(D_0) = 0$. Since $\Phi(D_i) \ge 0$ for all *i*, the total amortized cost of a sequence of *n* INCREMENT operations is an upper bound on the total actual cost, and so the worst-case cost of *n* INCREMENT operations is O(n).

The potential method gives us an easy way to analyze the counter even when it does not start at zero. The counter starts with b_0 1s, and after *n* INCREMENT

operations it has b_n 1s, where $0 \le b_0, b_n \le k$. (Recall that k is the number of bits in the counter.) We can rewrite equation (17.3) as

$$\sum_{i=1}^{n} c_i = \sum_{i=1}^{n} \hat{c}_i - \Phi(D_n) + \Phi(D_0) .$$
(17.4)

We have $\hat{c}_i \leq 2$ for all $1 \leq i \leq n$. Since $\Phi(D_0) = b_0$ and $\Phi(D_n) = b_n$, the total actual cost of *n* INCREMENT operations is

$$\sum_{i=1}^{n} c_i \leq \sum_{i=1}^{n} 2 - b_n + b_0$$

= $2n - b_n + b_0$.

Note in particular that since $b_0 \leq k$, as long as k = O(n), the total actual cost is O(n). In other words, if we execute at least $n = \Omega(k)$ INCREMENT operations, the total actual cost is O(n), no matter what initial value the counter contains.

Exercises

17.3-1

Suppose we have a potential function Φ such that $\Phi(D_i) \ge \Phi(D_0)$ for all *i*, but $\Phi(D_0) \ne 0$. Show that there exists a potential function Φ' such that $\Phi'(D_0) = 0$, $\Phi'(D_i) \ge 0$ for all $i \ge 1$, and the amortized costs using Φ' are the same as the amortized costs using Φ .

17.3-2

Redo Exercise 17.1-3 using a potential method of analysis.

17.3-3

Consider an ordinary binary min-heap data structure with *n* elements supporting the instructions INSERT and EXTRACT-MIN in $O(\lg n)$ worst-case time. Give a potential function Φ such that the amortized cost of INSERT is $O(\lg n)$ and the amortized cost of EXTRACT-MIN is O(1), and show that it works.

17.3**-**4

What is the total cost of executing n of the stack operations PUSH, POP, and MULTIPOP, assuming that the stack begins with s_0 objects and finishes with s_n objects?

17.3-5

Suppose that a counter begins at a number with b 1s in its binary representation, rather than at 0. Show that the cost of performing n INCREMENT operations is O(n) if $n = \Omega(b)$. (Do not assume that b is constant.)

17.3-6

Show how to implement a queue with two ordinary stacks (Exercise 10.1-6) so that the amortized cost of each ENQUEUE and each DEQUEUE operation is O(1).

17.3-7

Design a data structure to support the following two operations for a dynamic multiset S of integers, which allows duplicate values:

INSERT(S, x) inserts x into S.

DELETE-LARGER-HALF(S) deletes the largest $\lceil |S|/2 \rceil$ elements from S.

Explain how to implement this data structure so that any sequence of m INSERT and DELETE-LARGER-HALF operations runs in O(m) time. Your implementation should also include a way to output the elements of S in O(|S|) time.

17.4 Dynamic tables

We do not always know in advance how many objects some applications will store in a table. We might allocate space for a table, only to find out later that it is not enough. We must then reallocate the table with a larger size and copy all objects stored in the original table over into the new, larger table. Similarly, if many objects have been deleted from the table, it may be worthwhile to reallocate the table with a smaller size. In this section, we study this problem of dynamically expanding and contracting a table. Using amortized analysis, we shall show that the amortized cost of insertion and deletion is only O(1), even though the actual cost of an operation is large when it triggers an expansion or a contraction. Moreover, we shall see how to guarantee that the unused space in a dynamic table never exceeds a constant fraction of the total space.

We assume that the dynamic table supports the operations TABLE-INSERT and TABLE-DELETE. TABLE-INSERT inserts into the table an item that occupies a single *slot*, that is, a space for one item. Likewise, TABLE-DELETE removes an item from the table, thereby freeing a slot. The details of the data-structuring method used to organize the table are unimportant; we might use a stack (Section 10.1), a heap (Chapter 6), or a hash table (Chapter 11). We might also use an array or collection of arrays to implement object storage, as we did in Section 10.3.

We shall find it convenient to use a concept introduced in our analysis of hashing (Chapter 11). We define the *load factor* $\alpha(T)$ of a nonempty table *T* to be the number of items stored in the table divided by the size (number of slots) of the table. We assign an empty table (one with no items) size 0, and we define its load factor to be 1. If the load factor of a dynamic table is bounded below by a constant,



the unused space in the table is never more than a constant fraction of the total amount of space.

We start by analyzing a dynamic table in which we only insert items. We then consider the more general case in which we both insert and delete items.

17.4.1 Table expansion

Let us assume that storage for a table is allocated as an array of slots. A table fills up when all slots have been used or, equivalently, when its load factor is 1.¹ In some software environments, upon attempting to insert an item into a full table, the only alternative is to abort with an error. We shall assume, however, that our software environment, like many modern ones, provides a memory-management system that can allocate and free blocks of storage on request. Thus, upon inserting an item into a full table, we can *expand* the table by allocating a new table with more slots than the old table had. Because we always need the table to reside in contiguous memory, we must allocate a new array for the larger table and then copy items from the old table into the new table.

A common heuristic allocates a new table with twice as many slots as the old one. If the only table operations are insertions, then the load factor of the table is always at least 1/2, and thus the amount of wasted space never exceeds half the total space in the table.

In the following pseudocode, we assume that T is an object representing the table. The attribute *T.table* contains a pointer to the block of storage representing the table, *T.num* contains the number of items in the table, and *T.size* gives the total number of slots in the table. Initially, the table is empty: *T.num* = *T.size* = 0.

TABLE-INSERT(T, x)

- 1 **if** *T.size* == 0
- 2 allocate *T. table* with 1 slot
- 3 T.size = 1
- 4 **if** *T.num* == *T.size*
- 5 allocate *new-table* with $2 \cdot T$. *size* slots
- 6 insert all items in *T. table* into *new-table*
- 7 free *T.table*
- 8 T.table = new-table
- 9 $T.size = 2 \cdot T.size$
- 10 insert x into T.table
- 11 T.num = T.num + 1

 1 In some situations, such as an open-address hash table, we may wish to consider a table to be full if its load factor equals some constant strictly less than 1. (See Exercise 17.4-1.)

Notice that we have two "insertion" procedures here: the TABLE-INSERT procedure itself and the *elementary insertion* into a table in lines 6 and 10. We can analyze the running time of TABLE-INSERT in terms of the number of elementary insertions by assigning a cost of 1 to each elementary insertion. We assume that the actual running time of TABLE-INSERT is linear in the time to insert individual items, so that the overhead for allocating an initial table in line 2 is constant and the overhead for allocating and freeing storage in lines 5 and 7 is dominated by the cost of transferring items in line 6. We call the event in which lines 5–9 are executed an *expansion*.

Let us analyze a sequence of *n* TABLE-INSERT operations on an initially empty table. What is the cost c_i of the *i*th operation? If the current table has room for the new item (or if this is the first operation), then $c_i = 1$, since we need only perform the one elementary insertion in line 10. If the current table is full, however, and an expansion occurs, then $c_i = i$: the cost is 1 for the elementary insertion in line 10 plus i - 1 for the items that we must copy from the old table to the new table in line 6. If we perform *n* operations, the worst-case cost of an operation is O(n), which leads to an upper bound of $O(n^2)$ on the total running time for *n* operations.

This bound is not tight, because we rarely expand the table in the course of n TABLE-INSERT operations. Specifically, the *i*th operation causes an expansion only when i - 1 is an exact power of 2. The amortized cost of an operation is in fact O(1), as we can show using aggregate analysis. The cost of the *i*th operation is

$$c_i = \begin{cases} i & \text{if } i - 1 \text{ is an exact power of } 2, \\ 1 & \text{otherwise }. \end{cases}$$

The total cost of n TABLE-INSERT operations is therefore

$$\sum_{i=1}^{n} c_{i} \leq n + \sum_{j=0}^{\lfloor \lg n \rfloor} 2^{j}$$

< $n + 2n$
= $3n$,

because at most n operations cost 1 and the costs of the remaining operations form a geometric series. Since the total cost of n TABLE-INSERT operations is bounded by 3n, the amortized cost of a single operation is at most 3.

By using the accounting method, we can gain some feeling for why the amortized cost of a TABLE-INSERT operation should be 3. Intuitively, each item pays for 3 elementary insertions: inserting itself into the current table, moving itself when the table expands, and moving another item that has already been moved once when the table expands. For example, suppose that the size of the table is *m* immediately after an expansion. Then the table holds m/2 items, and it contains

no credit. We charge 3 dollars for each insertion. The elementary insertion that occurs immediately costs 1 dollar. We place another dollar as credit on the item inserted. We place the third dollar as credit on one of the m/2 items already in the table. The table will not fill again until we have inserted another m/2 - 1 items, and thus, by the time the table contains *m* items and is full, we will have placed a dollar on each item to pay to reinsert it during the expansion.

We can use the potential method to analyze a sequence of *n* TABLE-INSERT operations, and we shall use it in Section 17.4.2 to design a TABLE-DELETE operation that has an O(1) amortized cost as well. We start by defining a potential function Φ that is 0 immediately after an expansion but builds to the table size by the time the table is full, so that we can pay for the next expansion by the potential. The function

$$\Phi(T) = 2 \cdot T.num - T.size \tag{17.5}$$

is one possibility. Immediately after an expansion, we have T.num = T.size/2, and thus $\Phi(T) = 0$, as desired. Immediately before an expansion, we have T.num = T.size, and thus $\Phi(T) = T.num$, as desired. The initial value of the potential is 0, and since the table is always at least half full, $T.num \ge T.size/2$, which implies that $\Phi(T)$ is always nonnegative. Thus, the sum of the amortized costs of *n* TABLE-INSERT operations gives an upper bound on the sum of the actual costs.

To analyze the amortized cost of the *i*th TABLE-INSERT operation, we let num_i denote the number of items stored in the table after the *i*th operation, $size_i$ denote the total size of the table after the *i*th operation, and Φ_i denote the potential after the *i*th operation. Initially, we have $num_0 = 0$, $size_0 = 0$, and $\Phi_0 = 0$.

If the *i*th TABLE-INSERT operation does not trigger an expansion, then we have $size_i = size_{i-1}$ and the amortized cost of the operation is

$$\hat{c}_i = c_i + \Phi_i - \Phi_{i-1} = 1 + (2 \cdot num_i - size_i) - (2 \cdot num_{i-1} - size_{i-1}) = 1 + (2 \cdot num_i - size_i) - (2(num_i - 1) - size_i) = 3.$$

If the *i*th operation does trigger an expansion, then we have $size_i = 2 \cdot size_{i-1}$ and $size_{i-1} = num_{i-1} = num_i - 1$, which implies that $size_i = 2 \cdot (num_i - 1)$. Thus, the amortized cost of the operation is

$$\hat{c}_i = c_i + \Phi_i - \Phi_{i-1} = num_i + (2 \cdot num_i - size_i) - (2 \cdot num_{i-1} - size_{i-1}) = num_i + (2 \cdot num_i - 2 \cdot (num_i - 1)) - (2(num_i - 1) - (num_i - 1)) = num_i + 2 - (num_i - 1) = 3.$$



Figure 17.3 The effect of a sequence of *n* TABLE-INSERT operations on the number num_i of items in the table, the number $size_i$ of slots in the table, and the potential $\Phi_i = 2 \cdot num_i - size_i$, each being measured after the *i*th operation. The thin line shows num_i , the dashed line shows $size_i$, and the thick line shows Φ_i . Notice that immediately before an expansion, the potential has built up to the number of items in the table, and therefore it can pay for moving all the items to the new table. Afterwards, the potential drops to 0, but it is immediately increased by 2 upon inserting the item that caused the expansion.

Figure 17.3 plots the values of num_i , $size_i$, and Φ_i against *i*. Notice how the potential builds to pay for expanding the table.

17.4.2 Table expansion and contraction

To implement a TABLE-DELETE operation, it is simple enough to remove the specified item from the table. In order to limit the amount of wasted space, however, we might wish to *contract* the table when the load factor becomes too small. Table contraction is analogous to table expansion: when the number of items in the table drops too low, we allocate a new, smaller table and then copy the items from the old table into the new one. We can then free the storage for the old table by returning it to the memory-management system. Ideally, we would like to preserve two properties:

- the load factor of the dynamic table is bounded below by a positive constant, and
- the amortized cost of a table operation is bounded above by a constant.

We assume that we measure the cost in terms of elementary insertions and deletions.

You might think that we should double the table size upon inserting an item into a full table and halve the size when a deleting an item would cause the table to become less than half full. This strategy would guarantee that the load factor of the table never drops below 1/2, but unfortunately, it can cause the amortized cost of an operation to be quite large. Consider the following scenario. We perform *n* operations on a table *T*, where *n* is an exact power of 2. The first n/2 operations are insertions, which by our previous analysis cost a total of $\Theta(n)$. At the end of this sequence of insertions, T.num = T.size = n/2. For the second n/2 operations, we perform the following sequence:

insert, delete, delete, insert, insert, delete, delete, insert, insert,

The first insertion causes the table to expand to size n. The two following deletions cause the table to contract back to size n/2. Two further insertions cause another expansion, and so forth. The cost of each expansion and contraction is $\Theta(n)$, and there are $\Theta(n)$ of them. Thus, the total cost of the n operations is $\Theta(n^2)$, making the amortized cost of an operation $\Theta(n)$.

The downside of this strategy is obvious: after expanding the table, we do not delete enough items to pay for a contraction. Likewise, after contracting the table, we do not insert enough items to pay for an expansion.

We can improve upon this strategy by allowing the load factor of the table to drop below 1/2. Specifically, we continue to double the table size upon inserting an item into a full table, but we halve the table size when deleting an item causes the table to become less than 1/4 full, rather than 1/2 full as before. The load factor of the table is therefore bounded below by the constant 1/4.

Intuitively, we would consider a load factor of 1/2 to be ideal, and the table's potential would then be 0. As the load factor deviates from 1/2, the potential increases so that by the time we expand or contract the table, the table has garnered sufficient potential to pay for copying all the items into the newly allocated table. Thus, we will need a potential function that has grown to *T.num* by the time that the load factor has either increased to 1 or decreased to 1/4. After either expanding or contracting the table, the load factor goes back to 1/2 and the table's potential reduces back to 0.

We omit the code for TABLE-DELETE, since it is analogous to TABLE-INSERT. For our analysis, we shall assume that whenever the number of items in the table drops to 0, we free the storage for the table. That is, if T.num = 0, then T.size = 0.

We can now use the potential method to analyze the cost of a sequence of n TABLE-INSERT and TABLE-DELETE operations. We start by defining a potential function Φ that is 0 immediately after an expansion or contraction and builds as the load factor increases to 1 or decreases to 1/4. Let us denote the load fac-



Figure 17.4 The effect of a sequence of *n* TABLE-INSERT and TABLE-DELETE operations on the number num_i of items in the table, the number $size_i$ of slots in the table, and the potential

$$\Phi_i = \begin{cases} 2 \cdot num_i - size_i & \text{if } \alpha_i \ge 1/2 ,\\ size_i/2 - num_i & \text{if } \alpha_i < 1/2 , \end{cases}$$

each measured after the *i*th operation. The thin line shows num_i , the dashed line shows $size_i$, and the thick line shows Φ_i . Notice that immediately before an expansion, the potential has built up to the number of items in the table, and therefore it can pay for moving all the items to the new table. Likewise, immediately before a contraction, the potential has built up to the number of items in the table.

tor of a nonempty table T by $\alpha(T) = T.num/T.size$. Since for an empty table, T.num = T.size = 0 and $\alpha(T) = 1$, we always have T.num = $\alpha(T) \cdot T.size$, whether the table is empty or not. We shall use as our potential function

$$\Phi(T) = \begin{cases} 2 \cdot T.num - T.size & \text{if } \alpha(T) \ge 1/2, \\ T.size/2 - T.num & \text{if } \alpha(T) < 1/2. \end{cases}$$
(17.6)

Observe that the potential of an empty table is 0 and that the potential is never negative. Thus, the total amortized cost of a sequence of operations with respect to Φ provides an upper bound on the actual cost of the sequence.

Before proceeding with a precise analysis, we pause to observe some properties of the potential function, as illustrated in Figure 17.4. Notice that when the load factor is 1/2, the potential is 0. When the load factor is 1, we have T.size = T.num, which implies $\Phi(T) = T.num$, and thus the potential can pay for an expansion if an item is inserted. When the load factor is 1/4, we have $T.size = 4 \cdot T.num$, which

implies $\Phi(T) = T.num$, and thus the potential can pay for a contraction if an item is deleted.

To analyze a sequence of *n* TABLE-INSERT and TABLE-DELETE operations, we let c_i denote the actual cost of the *i*th operation, \hat{c}_i denote its amortized cost with respect to Φ , *num_i* denote the number of items stored in the table after the *i*th operation, *size_i* denote the total size of the table after the *i*th operation, α_i denote the load factor of the table after the *i*th operation, and Φ_i denote the potential after the *i*th operation. Initially, *num*₀ = 0, *size*₀ = 0, $\alpha_0 = 1$, and $\Phi_0 = 0$.

We start with the case in which the *i*th operation is TABLE-INSERT. The analysis is identical to that for table expansion in Section 17.4.1 if $\alpha_{i-1} \ge 1/2$. Whether the table expands or not, the amortized cost \hat{c}_i of the operation is at most 3. If $\alpha_{i-1} < 1/2$, the table cannot expand as a result of the operation, since the table expands only when $\alpha_{i-1} = 1$. If $\alpha_i < 1/2$ as well, then the amortized cost of the *i*th operation is

$$\hat{c}_i = c_i + \Phi_i - \Phi_{i-1} = 1 + (size_i/2 - num_i) - (size_{i-1}/2 - num_{i-1}) = 1 + (size_i/2 - num_i) - (size_i/2 - (num_i - 1)) = 0.$$

If $\alpha_{i-1} < 1/2$ but $\alpha_i \ge 1/2$, then

$$\begin{aligned} \hat{c}_{i} &= c_{i} + \Phi_{i} - \Phi_{i-1} \\ &= 1 + (2 \cdot num_{i} - size_{i}) - (size_{i-1}/2 - num_{i-1}) \\ &= 1 + (2(num_{i-1} + 1) - size_{i-1}) - (size_{i-1}/2 - num_{i-1}) \\ &= 3 \cdot num_{i-1} - \frac{3}{2}size_{i-1} + 3 \\ &= 3\alpha_{i-1}size_{i-1} - \frac{3}{2}size_{i-1} + 3 \\ &< \frac{3}{2}size_{i-1} - \frac{3}{2}size_{i-1} + 3 \\ &= 3 \end{aligned}$$

Thus, the amortized cost of a TABLE-INSERT operation is at most 3.

We now turn to the case in which the *i*th operation is TABLE-DELETE. In this case, $num_i = num_{i-1} - 1$. If $\alpha_{i-1} < 1/2$, then we must consider whether the operation causes the table to contract. If it does not, then $size_i = size_{i-1}$ and the amortized cost of the operation is

$$\hat{c}_i = c_i + \Phi_i - \Phi_{i-1} = 1 + (size_i/2 - num_i) - (size_{i-1}/2 - num_{i-1}) = 1 + (size_i/2 - num_i) - (size_i/2 - (num_i + 1)) = 2.$$

If $\alpha_{i-1} < 1/2$ and the *i*th operation does trigger a contraction, then the actual cost of the operation is $c_i = num_i + 1$, since we delete one item and move num_i items. We have $size_i/2 = size_{i-1}/4 = num_{i-1} = num_i + 1$, and the amortized cost of the operation is

$$\widehat{c}_i = c_i + \Phi_i - \Phi_{i-1} = (num_i + 1) + (size_i/2 - num_i) - (size_{i-1}/2 - num_{i-1}) = (num_i + 1) + ((num_i + 1) - num_i) - ((2 \cdot num_i + 2) - (num_i + 1)) = 1.$$

When the *i*th operation is a TABLE-DELETE and $\alpha_{i-1} \ge 1/2$, the amortized cost is also bounded above by a constant. We leave the analysis as Exercise 17.4-2.

In summary, since the amortized cost of each operation is bounded above by a constant, the actual time for any sequence of n operations on a dynamic table is O(n).

Exercises

17.4-1

Suppose that we wish to implement a dynamic, open-address hash table. Why might we consider the table to be full when its load factor reaches some value α that is strictly less than 1? Describe briefly how to make insertion into a dynamic, open-address hash table run in such a way that the expected value of the amortized cost per insertion is O(1). Why is the expected value of the actual cost per insertion not necessarily O(1) for all insertions?

17.4-2

Show that if $\alpha_{i-1} \ge 1/2$ and the *i*th operation on a dynamic table is TABLE-DELETE, then the amortized cost of the operation with respect to the potential function (17.6) is bounded above by a constant.

17.4-3

Suppose that instead of contracting a table by halving its size when its load factor drops below 1/4, we contract it by multiplying its size by 2/3 when its load factor drops below 1/3. Using the potential function

 $\Phi(T) = |2 \cdot T.num - T.size| ,$

show that the amortized cost of a TABLE-DELETE that uses this strategy is bounded above by a constant.

Problems

17-1 Bit-reversed binary counter

Chapter 30 examines an important algorithm called the fast Fourier transform, or FFT. The first step of the FFT algorithm performs a *bit-reversal permutation* on an input array A[0..n-1] whose length is $n = 2^k$ for some nonnegative integer k. This permutation swaps elements whose indices have binary representations that are the reverse of each other.

We can express each index a as a k-bit sequence $\langle a_{k-1}, a_{k-2}, \ldots, a_0 \rangle$, where $a = \sum_{i=0}^{k-1} a_i 2^i$. We define

$$\operatorname{rev}_k(\langle a_{k-1}, a_{k-2}, \dots, a_0 \rangle) = \langle a_0, a_1, \dots, a_{k-1} \rangle;$$

thus,

$$\operatorname{rev}_k(a) = \sum_{i=0}^{k-1} a_{k-i-1} 2^i$$

For example, if n = 16 (or, equivalently, k = 4), then $rev_k(3) = 12$, since the 4-bit representation of 3 is 0011, which when reversed gives 1100, the 4-bit representation of 12.

a. Given a function rev_k that runs in $\Theta(k)$ time, write an algorithm to perform the bit-reversal permutation on an array of length $n = 2^k$ in O(nk) time.

We can use an algorithm based on an amortized analysis to improve the running time of the bit-reversal permutation. We maintain a "bit-reversed counter" and a procedure BIT-REVERSED-INCREMENT that, when given a bit-reversed-counter value a, produces $rev_k(rev_k(a) + 1)$. If k = 4, for example, and the bit-reversed counter starts at 0, then successive calls to BIT-REVERSED-INCREMENT produce the sequence

 $0000, 1000, 0100, 1100, 0010, 1010, \ldots = 0, 8, 4, 12, 2, 10, \ldots$

- **b.** Assume that the words in your computer store k-bit values and that in unit time, your computer can manipulate the binary values with operations such as shifting left or right by arbitrary amounts, bitwise-AND, bitwise-OR, etc. Describe an implementation of the BIT-REVERSED-INCREMENT procedure that allows the bit-reversal permutation on an n-element array to be performed in a total of O(n) time.
- *c*. Suppose that you can shift a word left or right by only one bit in unit time. Is it still possible to implement an O(n)-time bit-reversal permutation?

17-2 Making binary search dynamic

Binary search of a sorted array takes logarithmic search time, but the time to insert a new element is linear in the size of the array. We can improve the time for insertion by keeping several sorted arrays.

Specifically, suppose that we wish to support SEARCH and INSERT on a set of *n* elements. Let $k = \lceil \lg(n+1) \rceil$, and let the binary representation of *n* be $\langle n_{k-1}, n_{k-2}, \ldots, n_0 \rangle$. We have *k* sorted arrays $A_0, A_1, \ldots, A_{k-1}$, where for $i = 0, 1, \ldots, k - 1$, the length of array A_i is 2^i . Each array is either full or empty, depending on whether $n_i = 1$ or $n_i = 0$, respectively. The total number of elements held in all *k* arrays is therefore $\sum_{i=0}^{k-1} n_i 2^i = n$. Although each individual array is sorted, elements in different arrays bear no particular relationship to each other.

- *a.* Describe how to perform the SEARCH operation for this data structure. Analyze its worst-case running time.
- **b.** Describe how to perform the INSERT operation. Analyze its worst-case and amortized running times.
- c. Discuss how to implement DELETE.

17-3 Amortized weight-balanced trees

Consider an ordinary binary search tree augmented by adding to each node x the attribute x.size giving the number of keys stored in the subtree rooted at x. Let α be a constant in the range $1/2 \le \alpha < 1$. We say that a given node x is α -balanced if x.left.size $\le \alpha \cdot x.size$ and x.right.size $\le \alpha \cdot x.size$. The tree as a whole is α -balanced if every node in the tree is α -balanced. The following amortized approach to maintaining weight-balanced trees was suggested by G. Varghese.

- *a.* A 1/2-balanced tree is, in a sense, as balanced as it can be. Given a node x in an arbitrary binary search tree, show how to rebuild the subtree rooted at x so that it becomes 1/2-balanced. Your algorithm should run in time $\Theta(x.size)$, and it can use O(x.size) auxiliary storage.
- **b.** Show that performing a search in an *n*-node α -balanced binary search tree takes $O(\lg n)$ worst-case time.

For the remainder of this problem, assume that the constant α is strictly greater than 1/2. Suppose that we implement INSERT and DELETE as usual for an *n*-node binary search tree, except that after every such operation, if any node in the tree is no longer α -balanced, then we "rebuild" the subtree rooted at the highest such node in the tree so that it becomes 1/2-balanced.

We shall analyze this rebuilding scheme using the potential method. For a node x in a binary search tree T, we define

$$\Delta(x) = |x.left.size - x.right.size|$$

and we define the potential of T as

$$\Phi(T) = c \sum_{x \in T: \Delta(x) \ge 2} \Delta(x) ,$$

where c is a sufficiently large constant that depends on α .

- *c.* Argue that any binary search tree has nonnegative potential and that a 1/2-balanced tree has potential 0.
- *d.* Suppose that *m* units of potential can pay for rebuilding an *m*-node subtree. How large must *c* be in terms of α in order for it to take O(1) amortized time to rebuild a subtree that is not α -balanced?
- *e.* Show that inserting a node into or deleting a node from an *n*-node α -balanced tree costs $O(\lg n)$ amortized time.

17-4 The cost of restructuring red-black trees

There are four basic operations on red-black trees that perform *structural modi-fications*: node insertions, node deletions, rotations, and color changes. We have seen that RB-INSERT and RB-DELETE use only O(1) rotations, node insertions, and node deletions to maintain the red-black properties, but they may make many more color changes.

a. Describe a legal red-black tree with *n* nodes such that calling RB-INSERT to add the (n + 1)st node causes $\Omega(\lg n)$ color changes. Then describe a legal red-black tree with *n* nodes for which calling RB-DELETE on a particular node causes $\Omega(\lg n)$ color changes.

Although the worst-case number of color changes per operation can be logarithmic, we shall prove that any sequence of m RB-INSERT and RB-DELETE operations on an initially empty red-black tree causes O(m) structural modifications in the worst case. Note that we count each color change as a structural modification.

b. Some of the cases handled by the main loop of the code of both RB-INSERT-FIXUP and RB-DELETE-FIXUP are *terminating*: once encountered, they cause the loop to terminate after a constant number of additional operations. For each of the cases of RB-INSERT-FIXUP and RB-DELETE-FIXUP, specify which are terminating and which are not. (*Hint:* Look at Figures 13.5, 13.6, and 13.7.)

We shall first analyze the structural modifications when only insertions are performed. Let T be a red-black tree, and define $\Phi(T)$ to be the number of red nodes in T. Assume that 1 unit of potential can pay for the structural modifications performed by any of the three cases of RB-INSERT-FIXUP.

- *c.* Let T' be the result of applying Case 1 of RB-INSERT-FIXUP to T. Argue that $\Phi(T') = \Phi(T) 1$.
- *d.* When we insert a node into a red-black tree using RB-INSERT, we can break the operation into three parts. List the structural modifications and potential changes resulting from lines 1–16 of RB-INSERT, from nonterminating cases of RB-INSERT-FIXUP, and from terminating cases of RB-INSERT-FIXUP.
- e. Using part (d), argue that the amortized number of structural modifications performed by any call of RB-INSERT is O(1).

We now wish to prove that there are O(m) structural modifications when there are both insertions and deletions. Let us define, for each node x,

 $w(x) = \begin{cases} 0 & \text{if } x \text{ is red }, \\ 1 & \text{if } x \text{ is black and has no red children }, \\ 0 & \text{if } x \text{ is black and has one red child }, \\ 2 & \text{if } x \text{ is black and has two red children }. \end{cases}$

Now we redefine the potential of a red-black tree T as

$$\Phi(T) = \sum_{x \in T} w(x) ,$$

and let T' be the tree that results from applying any nonterminating case of RB-INSERT-FIXUP or RB-DELETE-FIXUP to T.

- f. Show that $\Phi(T') \leq \Phi(T) 1$ for all nonterminating cases of RB-INSERT-FIXUP. Argue that the amortized number of structural modifications performed by any call of RB-INSERT-FIXUP is O(1).
- g. Show that $\Phi(T') \leq \Phi(T) 1$ for all nonterminating cases of RB-DELETE-FIXUP. Argue that the amortized number of structural modifications performed by any call of RB-DELETE-FIXUP is O(1).
- *h*. Complete the proof that in the worst case, any sequence of m RB-INSERT and RB-DELETE operations performs O(m) structural modifications.

17-5 Competitive analysis of self-organizing lists with move-to-front

A *self-organizing list* is a linked list of *n* elements, in which each element has a unique key. When we search for an element in the list, we are given a key, and we want to find an element with that key.

A self-organizing list has two important properties:

- 1. To find an element in the list, given its key, we must traverse the list from the beginning until we encounter the element with the given key. If that element is the *k*th element from the start of the list, then the cost to find the element is *k*.
- We may reorder the list elements after any operation, according to a given rule with a given cost. We may choose any heuristic we like to decide how to reorder the list.

Assume that we start with a given list of *n* elements, and we are given an access sequence $\sigma = \langle \sigma_1, \sigma_2, \dots, \sigma_m \rangle$ of keys to find, in order. The cost of the sequence is the sum of the costs of the individual accesses in the sequence.

Out of the various possible ways to reorder the list after an operation, this problem focuses on transposing adjacent list elements—switching their positions in the list—with a unit cost for each transpose operation. You will show, by means of a potential function, that a particular heuristic for reordering the list, move-to-front, entails a total cost no worse than 4 times that of any other heuristic for maintaining the list order—even if the other heuristic knows the access sequence in advance! We call this type of analysis a *competitive analysis*.

For a heuristic H and a given initial ordering of the list, denote the access cost of sequence σ by $C_H(\sigma)$. Let *m* be the number of accesses in σ .

a. Argue that if heuristic H does not know the access sequence in advance, then the worst-case cost for H on an access sequence σ is $C_H(\sigma) = \Omega(mn)$.

With the *move-to-front* heuristic, immediately after searching for an element *x*, we move *x* to the first position on the list (i.e., the front of the list).

Let rank_L(x) denote the rank of element x in list L, that is, the position of x in list L. For example, if x is the fourth element in L, then rank_L(x) = 4. Let c_i denote the cost of access σ_i using the move-to-front heuristic, which includes the cost of finding the element in the list and the cost of moving it to the front of the list by a series of transpositions of adjacent list elements.

b. Show that if σ_i accesses element x in list L using the move-to-front heuristic, then $c_i = 2 \cdot \operatorname{rank}_L(x) - 1$.

Now we compare move-to-front with any other heuristic H that processes an access sequence according to the two properties above. Heuristic H may transpose

elements in the list in any way it wants, and it might even know the entire access sequence in advance.

Let L_i be the list after access σ_i using move-to-front, and let L_i^* be the list after access σ_i using heuristic H. We denote the cost of access σ_i by c_i for move-to-front and by c_i^* for heuristic H. Suppose that heuristic H performs t_i^* transpositions during access σ_i .

c. In part (b), you showed that $c_i = 2 \cdot \operatorname{rank}_{L_{i-1}}(x) - 1$. Now show that $c_i^* = \operatorname{rank}_{L_{i-1}^*}(x) + t_i^*$.

We define an *inversion* in list L_i as a pair of elements y and z such that y precedes z in L_i and z precedes y in list L_i^* . Suppose that list L_i has q_i inversions after processing the access sequence $\langle \sigma_1, \sigma_2, \ldots, \sigma_i \rangle$. Then, we define a potential function Φ that maps L_i to a real number by $\Phi(L_i) = 2q_i$. For example, if L_i has the elements $\langle e, c, a, d, b \rangle$ and L_i^* has the elements $\langle c, a, b, d, e \rangle$, then L_i has 5 inversions ((e, c), (e, a), (e, d), (e, b), (d, b)), and so $\Phi(L_i) = 10$. Observe that $\Phi(L_i) \ge 0$ for all i and that, if move-to-front and heuristic H start with the same list L_0 , then $\Phi(L_0) = 0$.

d. Argue that a transposition either increases the potential by 2 or decreases the potential by 2.

Suppose that access σ_i finds the element *x*. To understand how the potential changes due to σ_i , let us partition the elements other than *x* into four sets, depending on where they are in the lists just before the *i*th access:

- Set A consists of elements that precede x in both L_{i-1} and L_{i-1}^* .
- Set *B* consists of elements that precede x in L_{i-1} and follow x in L_{i-1}^* .
- Set C consists of elements that follow x in L_{i-1} and precede x in L_{i-1}^* .
- Set D consists of elements that follow x in both L_{i-1} and L_{i-1}^* .
- *e.* Argue that $\operatorname{rank}_{L_{i-1}}(x) = |A| + |B| + 1$ and $\operatorname{rank}_{L_{i-1}^*}(x) = |A| + |C| + 1$.
- *f*. Show that access σ_i causes a change in potential of

 $\Phi(L_i) - \Phi(L_{i-1}) \le 2(|A| - |B| + t_i^*),$

where, as before, heuristic H performs t_i^* transpositions during access σ_i .

Define the amortized cost \hat{c}_i of access σ_i by $\hat{c}_i = c_i + \Phi(L_i) - \Phi(L_{i-1})$.

- **g.** Show that the amortized cost \hat{c}_i of access σ_i is bounded from above by $4c_i^*$.
- *h.* Conclude that the cost $C_{\text{MTF}}(\sigma)$ of access sequence σ with move-to-front is at most 4 times the cost $C_H(\sigma)$ of σ with any other heuristic H, assuming that both heuristics start with the same list.

Chapter notes

Aho, Hopcroft, and Ullman [5] used aggregate analysis to determine the running time of operations on a disjoint-set forest; we shall analyze this data structure using the potential method in Chapter 21. Tarjan [331] surveys the accounting and potential methods of amortized analysis and presents several applications. He attributes the accounting method to several authors, including M. R. Brown, R. E. Tarjan, S. Huddleston, and K. Mehlhorn. He attributes the potential method to D. D. Sleator. The term "amortized" is due to D. D. Sleator and R. E. Tarjan.

Potential functions are also useful for proving lower bounds for certain types of problems. For each configuration of the problem, we define a potential function that maps the configuration to a real number. Then we determine the potential Φ_{init} of the initial configuration, the potential Φ_{final} of the final configuration, and the maximum change in potential $\Delta \Phi_{max}$ due to any step. The number of steps must therefore be at least $|\Phi_{final} - \Phi_{init}| / |\Delta \Phi_{max}|$. Examples of potential functions to prove lower bounds in I/O complexity appear in works by Cormen, Sundquist, and Wisniewski [79]; Floyd [107]; and Aggarwal and Vitter [3]. Krumme, Cybenko, and Venkataraman [221] applied potential functions to prove lower bounds on *gossiping*: communicating a unique item from each vertex in a graph to every other vertex.

The move-to-front heuristic from Problem 17-5 works quite well in practice. Moreover, if we recognize that when we find an element, we can splice it out of its position in the list and relocate it to the front of the list in constant time, we can show that the cost of move-to-front is at most twice the cost of any other heuristic including, again, one that knows the entire access sequence in advance.

32 String Matching

Text-editing programs frequently need to find all occurrences of a pattern in the text. Typically, the text is a document being edited, and the pattern searched for is a particular word supplied by the user. Efficient algorithms for this problem—called "string matching"—can greatly aid the responsiveness of the text-editing program. Among their many other applications, string-matching algorithms search for particular patterns in DNA sequences. Internet search engines also use them to find Web pages relevant to queries.

We formalize the string-matching problem as follows. We assume that the text is an array T[1..n] of length n and that the pattern is an array P[1..m] of length $m \le n$. We further assume that the elements of P and T are characters drawn from a finite alphabet Σ . For example, we may have $\Sigma = \{0, 1\}$ or $\Sigma = \{a, b, ..., z\}$. The character arrays P and T are often called *strings* of characters.

Referring to Figure 32.1, we say that pattern *P* occurs with shift *s* in text *T* (or, equivalently, that pattern *P* occurs beginning at position s + 1 in text *T*) if $0 \le s \le n - m$ and T[s + 1 ... s + m] = P[1 ... m] (that is, if T[s + j] = P[j], for $1 \le j \le m$). If *P* occurs with shift *s* in *T*, then we call *s* a valid shift; otherwise, we call *s* an *invalid shift*. The string-matching problem is the problem of finding all valid shifts with which a given pattern *P* occurs in a given text *T*.



Figure 32.1 An example of the string-matching problem, where we want to find all occurrences of the pattern P = abaa in the text T = abcabaabcabac. The pattern occurs only once in the text, at shift s = 3, which we call a valid shift. A vertical line connects each character of the pattern to its matching character in the text, and all matched characters are shaded.

Algorithm	Preprocessing time	Matching time
Naive	0	O((n-m+1)m)
Rabin-Karp	$\Theta(m)$	O((n-m+1)m)
Finite automaton	$O(m \Sigma)$	$\Theta(n)$
Knuth-Morris-Pratt	$\Theta(m)$	$\Theta(n)$

Figure 32.2 The string-matching algorithms in this chapter and their preprocessing and matching times.

Except for the naive brute-force algorithm, which we review in Section 32.1, each string-matching algorithm in this chapter performs some preprocessing based on the pattern and then finds all valid shifts; we call this latter phase "matching." Figure 32.2 shows the preprocessing and matching times for each of the algorithms in this chapter. The total running time of each algorithm is the sum of the preprocessing and matching times. Section 32.2 presents an interesting string-matching algorithm, due to Rabin and Karp. Although the $\Theta((n - m + 1)m)$ worst-case running time of this algorithm is no better than that of the naive method, it works much better on average and in practice. It also generalizes nicely to other patternmatching problems. Section 32.3 then describes a string-matching algorithm that begins by constructing a finite automaton specifically designed to search for occurrences of the given pattern P in a text. This algorithm takes $O(m |\Sigma|)$ preprocessing time, but only $\Theta(n)$ matching time. Section 32.4 presents the similar, but much cleverer, Knuth-Morris-Pratt (or KMP) algorithm; it has the same $\Theta(n)$ matching time, and it reduces the preprocessing time to only $\Theta(m)$.

Notation and terminology

We denote by Σ^* (read "sigma-star") the set of all finite-length strings formed using characters from the alphabet Σ . In this chapter, we consider only finitelength strings. The zero-length *empty string*, denoted ε , also belongs to Σ^* . The length of a string x is denoted |x|. The *concatenation* of two strings x and y, denoted xy, has length |x| + |y| and consists of the characters from x followed by the characters from y.

We say that a string w is a **prefix** of a string x, denoted $w \sqsubset x$, if x = wy for some string $y \in \Sigma^*$. Note that if $w \sqsubset x$, then $|w| \le |x|$. Similarly, we say that a string w is a **suffix** of a string x, denoted $w \sqsupset x$, if x = yw for some $y \in \Sigma^*$. As with a prefix, $w \sqsupset x$ implies $|w| \le |x|$. For example, we have $ab \sqsubset abcca$ and $cca \sqsupset abcca$. The empty string ε is both a suffix and a prefix of every string. For any strings x and y and any character a, we have $x \sqsupset y$ if and only if $xa \sqsupset ya$.



Figure 32.3 A graphical proof of Lemma 32.1. We suppose that $x \Box z$ and $y \Box z$. The three parts of the figure illustrate the three cases of the lemma. Vertical lines connect matching regions (shown shaded) of the strings. (a) If $|x| \le |y|$, then $x \Box y$. (b) If $|x| \ge |y|$, then $y \Box x$. (c) If |x| = |y|, then x = y.

Also note that \square and \square are transitive relations. The following lemma will be useful later.

Lemma 32.1 (Overlapping-suffix lemma)

Suppose that x, y, and z are strings such that $x \sqsupset z$ and $y \sqsupset z$. If $|x| \le |y|$, then $x \sqsupset y$. If $|x| \ge |y|$, then $y \sqsupset x$. If |x| = |y|, then x = y.

Proof See Figure 32.3 for a graphical proof.

For brevity of notation, we denote the *k*-character prefix P[1..k] of the pattern P[1..m] by P_k . Thus, $P_0 = \varepsilon$ and $P_m = P = P[1..m]$. Similarly, we denote the *k*-character prefix of the text *T* by T_k . Using this notation, we can state the string-matching problem as that of finding all shifts *s* in the range $0 \le s \le n - m$ such that $P \supseteq T_{s+m}$.

In our pseudocode, we allow two equal-length strings to be compared for equality as a primitive operation. If the strings are compared from left to right and the comparison stops when a mismatch is discovered, we assume that the time taken by such a test is a linear function of the number of matching characters discovered. To be precise, the test "x == y" is assumed to take time $\Theta(t + 1)$, where t is the length of the longest string z such that $z \sqsubset x$ and $z \sqsubset y$. (We write $\Theta(t + 1)$ rather than $\Theta(t)$ to handle the case in which t = 0; the first characters compared do not match, but it takes a positive amount of time to perform this comparison.)

https://hemanthrajhemu.github.io

32.1 The naive string-matching algorithm

The naive algorithm finds all valid shifts using a loop that checks the condition P[1..m] = T[s + 1..s + m] for each of the n - m + 1 possible values of *s*.

NAIVE-STRING-MATCHER (T, P)

1 n = T.length2 m = P.length3 for s = 0 to n - m4 if $P[1 \dots m] == T[s + 1 \dots s + m]$ 5 print "Pattern occurs with shift" s

Figure 32.4 portrays the naive string-matching procedure as sliding a "template" containing the pattern over the text, noting for which shifts all of the characters on the template equal the corresponding characters in the text. The **for** loop of lines 3-5 considers each possible shift explicitly. The test in line 4 determines whether the current shift is valid; this test implicitly loops to check corresponding character positions until all positions match successfully or a mismatch is found. Line 5 prints out each valid shift *s*.

Procedure NAIVE-STRING-MATCHER takes time O((n - m + 1)m), and this bound is tight in the worst case. For example, consider the text string a^n (a string of *n* a's) and the pattern a^m . For each of the n - m + 1 possible values of the shift *s*, the implicit loop on line 4 to compare corresponding characters must execute *m* times to validate the shift. The worst-case running time is thus $\Theta((n - m + 1)m)$, which is $\Theta(n^2)$ if $m = \lfloor n/2 \rfloor$. Because it requires no preprocessing, NAIVE-STRING-MATCHER's running time equals its matching time.



Figure 32.4 The operation of the naive string matcher for the pattern P = aab and the text T = acaabc. We can imagine the pattern P as a template that we slide next to the text. (a)–(d) The four successive alignments tried by the naive string matcher. In each part, vertical lines connect corresponding regions found to match (shown shaded), and a jagged line connects the first mismatched character found, if any. The algorithm finds one occurrence of the pattern, at shift s = 2, shown in part (c).

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As we shall see, NAIVE-STRING-MATCHER is not an optimal procedure for this problem. Indeed, in this chapter we shall see that the Knuth-Morris-Pratt algorithm is much better in the worst case. The naive string-matcher is inefficient because it entirely ignores information gained about the text for one value of s when it considers other values of s. Such information can be quite valuable, however. For example, if P = aaab and we find that s = 0 is valid, then none of the shifts 1, 2, or 3 are valid, since T[4] = b. In the following sections, we examine several ways to make effective use of this sort of information.

Exercises

32.1-1

Show the comparisons the naive string matcher makes for the pattern P = 0001 in the text T = 000010001010001.

32.1-2

Suppose that all characters in the pattern P are different. Show how to accelerate NAIVE-STRING-MATCHER to run in time O(n) on an n-character text T.

32.1-3

Suppose that pattern *P* and text *T* are *randomly* chosen strings of length *m* and *n*, respectively, from the *d*-ary alphabet $\Sigma_d = \{0, 1, \dots, d-1\}$, where $d \ge 2$. Show that the *expected* number of character-to-character comparisons made by the implicit loop in line 4 of the naive algorithm is

$$(n-m+1)\frac{1-d^{-m}}{1-d^{-1}} \leq 2(n-m+1)$$

over all executions of this loop. (Assume that the naive algorithm stops comparing characters for a given shift once it finds a mismatch or matches the entire pattern.) Thus, for randomly chosen strings, the naive algorithm is quite efficient.

32.1-4

Suppose we allow the pattern P to contain occurrences of a *gap character* \diamond that can match an *arbitrary* string of characters (even one of zero length). For example, the pattern ab \diamond ba \diamond c occurs in the text cabccbacbacab as

 $\begin{array}{cccc} c & ab & cc & ba & cba & c & ab \\ \hline ab & \diamond & ba & \diamond & c \\ \hline and & as \\ \hline c & ab & ccbac & ba & c & c \\ \hline ab & \diamond & ba & \diamond & c \\ \hline \end{array}$

Note that the gap character may occur an arbitrary number of times in the pattern but not at all in the text. Give a polynomial-time algorithm to determine whether such a pattern P occurs in a given text T, and analyze the running time of your algorithm.

32.2 The Rabin-Karp algorithm

Rabin and Karp proposed a string-matching algorithm that performs well in practice and that also generalizes to other algorithms for related problems, such as two-dimensional pattern matching. The Rabin-Karp algorithm uses $\Theta(m)$ preprocessing time, and its worst-case running time is $\Theta((n-m+1)m)$. Based on certain assumptions, however, its average-case running time is better.

This algorithm makes use of elementary number-theoretic notions such as the equivalence of two numbers modulo a third number. You might want to refer to Section 31.1 for the relevant definitions.

For expository purposes, let us assume that $\Sigma = \{0, 1, 2, \dots, 9\}$, so that each character is a decimal digit. (In the general case, we can assume that each character is a digit in radix-d notation, where $d = |\Sigma|$.) We can then view a string of k consecutive characters as representing a length-k decimal number. The character string 31415 thus corresponds to the decimal number 31,415. Because we interpret the input characters as both graphical symbols and digits, we find it convenient in this section to denote them as we would digits, in our standard text font.

Given a pattern P[1..m], let p denote its corresponding decimal value. In a similar manner, given a text T[1..n], let t_s denote the decimal value of the length-m substring T[s + 1..s + m], for s = 0, 1, ..., n - m. Certainly, $t_s = p$ if and only if T[s + 1..s + m] = P[1..m]; thus, s is a valid shift if and only if $t_s = p$. If we could compute p in time $\Theta(m)$ and all the t_s values in a total of $\Theta(n-m+1)$ time,¹ then we could determine all valid shifts s in time $\Theta(m) + \Theta(n - m + 1) = \Theta(n)$ by comparing p with each of the t_s values. (For the moment, let's not worry about the possibility that p and the t_s values might be very large numbers.)

We can compute p in time $\Theta(m)$ using Horner's rule (see Section 30.1):

$$p = P[m] + 10 \left(P[m-1] + 10(P[m-2] + \dots + 10(P[2] + 10P[1]) \dots \right) \right)$$

Similarly, we can compute t_0 from $T[1 \dots m]$ in time $\Theta(m)$.

¹We write $\Theta(n - m + 1)$ instead of $\Theta(n - m)$ because *s* takes on n - m + 1 different values. The "+1" is significant in an asymptotic sense because when m = n, computing the lone t_s value takes $\Theta(1)$ time, not $\Theta(0)$ time.

To compute the remaining values $t_1, t_2, ..., t_{n-m}$ in time $\Theta(n-m)$, we observe that we can compute t_{s+1} from t_s in constant time, since

$$t_{s+1} = 10(t_s - 10^{m-1}T[s+1]) + T[s+m+1].$$
(32.1)

Subtracting $10^{m-1}T[s + 1]$ removes the high-order digit from t_s , multiplying the result by 10 shifts the number left by one digit position, and adding T[s + m + 1] brings in the appropriate low-order digit. For example, if m = 5 and $t_s = 31415$, then we wish to remove the high-order digit T[s + 1] = 3 and bring in the new low-order digit (suppose it is T[s + 5 + 1] = 2) to obtain

$$t_{s+1} = 10(31415 - 10000 \cdot 3) + 2$$

= 14152.

If we precompute the constant 10^{m-1} (which we can do in time $O(\lg m)$ using the techniques of Section 31.6, although for this application a straightforward O(m)-time method suffices), then each execution of equation (32.1) takes a constant number of arithmetic operations. Thus, we can compute p in time $\Theta(m)$, and we can compute all of $t_0, t_1, \ldots, t_{n-m}$ in time $\Theta(n-m+1)$. Therefore, we can find all occurrences of the pattern $P[1 \ldots m]$ in the text $T[1 \ldots n]$ with $\Theta(m)$ preprocessing time and $\Theta(n-m+1)$ matching time.

Until now, we have intentionally overlooked one problem: p and t_s may be too large to work with conveniently. If P contains m characters, then we cannot reasonably assume that each arithmetic operation on p (which is m digits long) takes "constant time." Fortunately, we can solve this problem easily, as Figure 32.5 shows: compute p and the t_s values modulo a suitable modulus q. We can compute p modulo q in $\Theta(m)$ time and all the t_s values modulo q in $\Theta(n - m + 1)$ time. If we choose the modulus q as a prime such that 10q just fits within one computer word, then we can perform all the necessary computations with single-precision arithmetic. In general, with a d-ary alphabet $\{0, 1, \ldots, d - 1\}$, we choose q so that dq fits within a computer word and adjust the recurrence equation (32.1) to work modulo q, so that it becomes

$$t_{s+1} = (d(t_s - T[s+1]h) + T[s+m+1]) \mod q , \qquad (32.2)$$

where $h \equiv d^{m-1} \pmod{q}$ is the value of the digit "1" in the high-order position of an *m*-digit text window.

The solution of working modulo q is not perfect, however: $t_s \equiv p \pmod{q}$ does not imply that $t_s \equiv p$. On the other hand, if $t_s \not\equiv p \pmod{q}$, then we definitely have that $t_s \neq p$, so that shift s is invalid. We can thus use the test $t_s \equiv p \pmod{q}$ as a fast heuristic test to rule out invalid shifts s. Any shift s for which $t_s \equiv p \pmod{q}$ must be tested further to see whether s is really valid or we just have a **spurious hit**. This additional test explicitly checks the condition



Figure 32.5 The Rabin-Karp algorithm. Each character is a decimal digit, and we compute values modulo 13. (a) A text string. A window of length 5 is shown shaded. The numerical value of the shaded number, computed modulo 13, yields the value 7. (b) The same text string with values computed modulo 13 for each possible position of a length-5 window. Assuming the pattern P = 31415, we look for windows whose value modulo 13 is 7, since $31415 \equiv 7 \pmod{13}$. The algorithm finds two such windows, shown shaded in the figure. The first, beginning at text position 7, is indeed an occurrence of the pattern, while the second, beginning at text position 13, is a spurious hit. (c) How to compute the value for a window in constant time, given the value for the previous window. The first window has value 31415. Dropping the high-order digit 3, shifting left (multiplying by 10), and then adding in the low-order digit 2 gives us the new value 14152. Because all computations are performed modulo 13, the value for the first window is 7, and the value for the new window is 8.

P[1..m] = T[s + 1..s + m]. If q is large enough, then we hope that spurious hits occur infrequently enough that the cost of the extra checking is low.

The following procedure makes these ideas precise. The inputs to the procedure are the text T, the pattern P, the radix d to use (which is typically taken to be $|\Sigma|$), and the prime q to use.

RABIN-KARP-MATCHER (T, P, d, q)

1 n = T.length2 m = P.length $h = d^{m-1} \mod q$ 3 p = 04 5 $t_0 = 0$ 6 for i = 1 to m // preprocessing $p = (dp + P[i]) \mod q$ 7 8 $t_0 = (dt_0 + T[i]) \mod q$ 9 for s = 0 to n - m// matching 10 if $p == t_s$ 11 **if** P[1...m] == T[s + 1...s + m]12 print "Pattern occurs with shift" s 13 if s < n - m $t_{s+1} = (d(t_s - T[s+1]h) + T[s+m+1]) \mod q$ 14

The procedure RABIN-KARP-MATCHER works as follows. All characters are interpreted as radix-*d* digits. The subscripts on *t* are provided only for clarity; the program works correctly if all the subscripts are dropped. Line 3 initializes *h* to the value of the high-order digit position of an *m*-digit window. Lines 4–8 compute *p* as the value of $P[1..m] \mod q$ and t_0 as the value of $T[1..m] \mod q$. The for loop of lines 9–14 iterates through all possible shifts *s*, maintaining the following invariant:

Whenever line 10 is executed, $t_s = T[s + 1 \dots s + m] \mod q$.

If $p = t_s$ in line 10 (a "hit"), then line 11 checks to see whether P[1..m] = T[s+1..s+m] in order to rule out the possibility of a spurious hit. Line 12 prints out any valid shifts that are found. If s < n - m (checked in line 13), then the **for** loop will execute at least one more time, and so line 14 first executes to ensure that the loop invariant holds when we get back to line 10. Line 14 computes the value of $t_{s+1} \mod q$ from the value of $t_s \mod q$ in constant time using equation (32.2) directly.

RABIN-KARP-MATCHER takes $\Theta(m)$ preprocessing time, and its matching time is $\Theta((n - m + 1)m)$ in the worst case, since (like the naive string-matching algorithm) the Rabin-Karp algorithm explicitly verifies every valid shift. If $P = a^m$

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and $T = a^n$, then verifying takes time $\Theta((n-m+1)m)$, since each of the n-m+1 possible shifts is valid.

In many applications, we expect few valid shifts—perhaps some constant c of them. In such applications, the expected matching time of the algorithm is only O((n - m + 1) + cm) = O(n + m), plus the time required to process spurious hits. We can base a heuristic analysis on the assumption that reducing values modulo q acts like a random mapping from Σ^* to \mathbb{Z}_q . (See the discussion on the use of division for hashing in Section 11.3.1. It is difficult to formalize and prove such an assumption, although one viable approach is to assume that q is chosen randomly from integers of the appropriate size. We shall not pursue this formalization here.) We can then expect that the number of spurious hits is O(n/q), since we can estimate the chance that an arbitrary t_s will be equivalent to p, modulo q, as 1/q. Since there are O(n) positions at which the test of line 10 fails and we spend O(m) time for each hit, the expected matching time taken by the Rabin-Karp algorithm is

 $O(n) + O(m(\nu + n/q))$,

where v is the number of valid shifts. This running time is O(n) if v = O(1) and we choose $q \ge m$. That is, if the expected number of valid shifts is small (O(1)) and we choose the prime q to be larger than the length of the pattern, then we can expect the Rabin-Karp procedure to use only O(n + m) matching time. Since $m \le n$, this expected matching time is O(n).

Exercises

32.2-1

Working modulo q = 11, how many spurious hits does the Rabin-Karp matcher encounter in the text T = 3141592653589793 when looking for the pattern P = 26?

32.2-2

How would you extend the Rabin-Karp method to the problem of searching a text string for an occurrence of any one of a given set of k patterns? Start by assuming that all k patterns have the same length. Then generalize your solution to allow the patterns to have different lengths.

32.2-3

Show how to extend the Rabin-Karp method to handle the problem of looking for a given $m \times m$ pattern in an $n \times n$ array of characters. (The pattern may be shifted vertically and horizontally, but it may not be rotated.)

32.2-4

Alice has a copy of a long *n*-bit file $A = \langle a_{n-1}, a_{n-2}, \ldots, a_0 \rangle$, and Bob similarly has an *n*-bit file $B = \langle b_{n-1}, b_{n-2}, \ldots, b_0 \rangle$. Alice and Bob wish to know if their files are identical. To avoid transmitting all of *A* or *B*, they use the following fast probabilistic check. Together, they select a prime q > 1000n and randomly select an integer *x* from $\{0, 1, \ldots, q-1\}$. Then, Alice evaluates

$$A(x) = \left(\sum_{i=0}^{n-1} a_i x^i\right) \mod q$$

and Bob similarly evaluates B(x). Prove that if $A \neq B$, there is at most one chance in 1000 that A(x) = B(x), whereas if the two files are the same, A(x) is necessarily the same as B(x). (*Hint:* See Exercise 31.4-4.)

32.3 String matching with finite automata

Many string-matching algorithms build a finite automaton—a simple machine for processing information—that scans the text string T for all occurrences of the pattern P. This section presents a method for building such an automaton. These string-matching automata are very efficient: they examine each text character *exactly once*, taking constant time per text character. The matching time used—after preprocessing the pattern to build the automaton—is therefore $\Theta(n)$. The time to build the automaton, however, can be large if Σ is large. Section 32.4 describes a clever way around this problem.

We begin this section with the definition of a finite automaton. We then examine a special string-matching automaton and show how to use it to find occurrences of a pattern in a text. Finally, we shall show how to construct the string-matching automaton for a given input pattern.

Finite automata

A *finite automaton* M, illustrated in Figure 32.6, is a 5-tuple $(Q, q_0, A, \Sigma, \delta)$, where

- *Q* is a finite set of *states*,
- $q_0 \in Q$ is the *start state*,
- $A \subseteq Q$ is a distinguished set of *accepting states*,
- Σ is a finite *input alphabet*,
- δ is a function from $Q \times \Sigma$ into Q, called the *transition function* of M.



Figure 32.6 A simple two-state finite automaton with state set $Q = \{0, 1\}$, start state $q_0 = 0$, and input alphabet $\Sigma = \{a, b\}$. (a) A tabular representation of the transition function δ . (b) An equivalent state-transition diagram. State 1, shown blackend, is the only accepting state. Directed edges represent transitions. For example, the edge from state 1 to state 0 labeled b indicates that $\delta(1, b) = 0$. This automaton accepts those strings that end in an odd number of a's. More precisely, it accepts a string x if and only if x = yz, where $y = \varepsilon$ or y ends with a b, and $z = a^k$, where k is odd. For example, on input abaaa, including the start state, this automaton enters the sequence of states $\langle 0, 1, 0, 1, 0, 1 \rangle$, and so it accepts this input. For input abbaa, it enters the sequence of states $\langle 0, 1, 0, 0, 1, 0 \rangle$, and so it rejects this input.

The finite automaton begins in state q_0 and reads the characters of its input string one at a time. If the automaton is in state q and reads input character a, it moves ("makes a transition") from state q to state $\delta(q, a)$. Whenever its current state q is a member of A, the machine M has **accepted** the string read so far. An input that is not accepted is **rejected**.

A finite automaton M induces a function ϕ , called the *final-state function*, from Σ^* to Q such that $\phi(w)$ is the state M ends up in after scanning the string w. Thus, M accepts a string w if and only if $\phi(w) \in A$. We define the function ϕ recursively, using the transition function:

$$\begin{split} \phi(\varepsilon) &= q_0 , \\ \phi(wa) &= \delta(\phi(w), a) \quad \text{for } w \in \Sigma^*, a \in \Sigma . \end{split}$$

String-matching automata

For a given pattern P, we construct a string-matching automaton in a preprocessing step before using it to search the text string. Figure 32.7 illustrates how we construct the automaton for the pattern P = ababaca. From now on, we shall assume that P is a given fixed pattern string; for brevity, we shall not indicate the dependence upon P in our notation.

In order to specify the string-matching automaton corresponding to a given pattern P[1..m], we first define an auxiliary function σ , called the *suffix function* corresponding to P. The function σ maps Σ^* to $\{0, 1, ..., m\}$ such that $\sigma(x)$ is the length of the longest prefix of P that is also a suffix of x:

$$\sigma(x) = \max\left\{k : P_k \sqsupset x\right\} .$$

(32.3)


Figure 32.7 (a) A state-transition diagram for the string-matching automaton that accepts all strings ending in the string ababaca. State 0 is the start state, and state 7 (shown blackened) is the only accepting state. A directed edge from state *i* to state *j* labeled *a* represents $\delta(i, a) = j$. The right-going edges forming the "spine" of the automaton, shown heavy in the figure, correspond to successful matches between pattern and input characters. The left-going edges correspond to failing matches. Some edges corresponding to failing matches are omitted; by convention, if a state *i* has no outgoing edge labeled *a* for some $a \in \Sigma$, then $\delta(i, a) = 0$. (b) The corresponding transition function δ , and the pattern string P = ababaca. The entries corresponding to successful matches between pattern and input characters are shown shaded. (c) The operation of the automaton on the text T = abababacaba. Under each text character T[i] appears the state $\phi(T_i)$ that the automaton is in after processing the prefix T_i . The automaton finds one occurrence of the pattern, ending in position 9.

The suffix function σ is well defined since the empty string $P_0 = \varepsilon$ is a suffix of every string. As examples, for the pattern P = ab, we have $\sigma(\varepsilon) = 0$, $\sigma(\texttt{ccaca}) = 1$, and $\sigma(\texttt{ccab}) = 2$. For a pattern P of length m, we have $\sigma(x) = m$ if and only if $P \sqsupset x$. From the definition of the suffix function, $x \sqsupset y$ implies $\sigma(x) \le \sigma(y)$.

We define the string-matching automaton that corresponds to a given pattern P[1..m] as follows:

- The state set Q is $\{0, 1, ..., m\}$. The start state q_0 is state 0, and state m is the only accepting state.
- The transition function δ is defined by the following equation, for any state q and character a:

$$\delta(q,a) = \sigma(P_q a) . \tag{32.4}$$

We define $\delta(q, a) = \sigma(P_q a)$ because we want to keep track of the longest prefix of the pattern P that has matched the text string T so far. We consider the most recently read characters of T. In order for a substring of T—let's say the substring ending at T[i]—to match some prefix P_j of P, this prefix P_j must be a suffix of T_i . Suppose that $q = \phi(T_i)$, so that after reading T_i , the automaton is in state q. We design the transition function δ so that this state number, q, tells us the length of the longest prefix of P that matches a suffix of T_i . That is, in state q, $P_q \Box T_i$ and $q = \sigma(T_i)$. (Whenever q = m, all m characters of P match a suffix of T_i , and so we have found a match.) Thus, since $\phi(T_i)$ and $\sigma(T_i)$ both equal q, we shall see (in Theorem 32.4, below) that the automaton maintains the following invariant:

$$\phi(T_i) = \sigma(T_i) . \tag{32.5}$$

If the automaton is in state q and reads the next character T[i + 1] = a, then we want the transition to lead to the state corresponding to the longest prefix of P that is a suffix of $T_i a$, and that state is $\sigma(T_i a)$. Because P_q is the longest prefix of P that is a suffix of T_i , the longest prefix of P that is a suffix of $T_i a$ is not only $\sigma(T_i a)$, but also $\sigma(P_q a)$. (Lemma 32.3, on page 1000, proves that $\sigma(T_i a) = \sigma(P_q a)$.) Thus, when the automaton is in state q, we want the transition function on character a to take the automaton to state $\sigma(P_q a)$.

There are two cases to consider. In the first case, a = P[q + 1], so that the character *a* continues to match the pattern; in this case, because $\delta(q, a) = q+1$, the transition continues to go along the "spine" of the automaton (the heavy edges in Figure 32.7). In the second case, $a \neq P[q+1]$, so that *a* does not continue to match the pattern. Here, we must find a smaller prefix of *P* that is also a suffix of T_i . Because the preprocessing step matches the pattern against itself when creating the string-matching automaton, the transition function quickly identifies the longest such smaller prefix of *P*.

Let's look at an example. The string-matching automaton of Figure 32.7 has $\delta(5, c) = 6$, illustrating the first case, in which the match continues. To illustrate the second case, observe that the automaton of Figure 32.7 has $\delta(5, b) = 4$. We make this transition because if the automaton reads a b in state q = 5, then $P_q b = ababab$, and the longest prefix of P that is also a suffix of ababab is $P_4 = abab$.



Figure 32.8 An illustration for the proof of Lemma 32.2. The figure shows that $r \le \sigma(x) + 1$, where $r = \sigma(xa)$.

To clarify the operation of a string-matching automaton, we now give a simple, efficient program for simulating the behavior of such an automaton (represented by its transition function δ) in finding occurrences of a pattern *P* of length *m* in an input text T[1..n]. As for any string-matching automaton for a pattern of length *m*, the state set *Q* is $\{0, 1, ..., m\}$, the start state is 0, and the only accepting state is state *m*.

FINITE-AUTOMATON-MATCHER (T, δ, m)

1 n = T.length2 q = 03 for i = 1 to n4 $q = \delta(q, T[i])$ 5 if q == m6 print "Pattern occurs with shift" i - m

From the simple loop structure of FINITE-AUTOMATON-MATCHER, we can easily see that its matching time on a text string of length n is $\Theta(n)$. This matching time, however, does not include the preprocessing time required to compute the transition function δ . We address this problem later, after first proving that the procedure FINITE-AUTOMATON-MATCHER operates correctly.

Consider how the automaton operates on an input text T[1..n]. We shall prove that the automaton is in state $\sigma(T_i)$ after scanning character T[i]. Since $\sigma(T_i) = m$ if and only if $P \supseteq T_i$, the machine is in the accepting state *m* if and only if it has just scanned the pattern *P*. To prove this result, we make use of the following two lemmas about the suffix function σ .

Lemma 32.2 (Suffix-function inequality)

For any string x and character a, we have $\sigma(xa) \le \sigma(x) + 1$.

Proof Referring to Figure 32.8, let $r = \sigma(xa)$. If r = 0, then the conclusion $\sigma(xa) = r \le \sigma(x) + 1$ is trivially satisfied, by the nonnegativity of $\sigma(x)$. Now assume that r > 0. Then, $P_r \sqsupset xa$, by the definition of σ . Thus, $P_{r-1} \sqsupset x$, by



Figure 32.9 An illustration for the proof of Lemma 32.3. The figure shows that $r = \sigma(P_q a)$, where $q = \sigma(x)$ and $r = \sigma(xa)$.

dropping the *a* from the end of P_r and from the end of xa. Therefore, $r-1 \le \sigma(x)$, since $\sigma(x)$ is the largest *k* such that $P_k \supseteq x$, and thus $\sigma(xa) = r \le \sigma(x) + 1$.

Lemma 32.3 (Suffix-function recursion lemma)

For any string x and character a, if $q = \sigma(x)$, then $\sigma(xa) = \sigma(P_qa)$.

Proof From the definition of σ , we have $P_q \Box x$. As Figure 32.9 shows, we also have $P_q a \Box xa$. If we let $r = \sigma(xa)$, then $P_r \Box xa$ and, by Lemma 32.2, $r \le q + 1$. Thus, we have $|P_r| = r \le q + 1 = |P_q a|$. Since $P_q a \Box xa$, $P_r \Box xa$, and $|P_r| \le |P_q a|$, Lemma 32.1 implies that $P_r \Box P_q a$. Therefore, $r \le \sigma(P_q a)$, that is, $\sigma(xa) \le \sigma(P_q a)$. But we also have $\sigma(P_q a) \le \sigma(xa)$, since $P_q a \Box xa$. Thus, $\sigma(xa) = \sigma(P_q a)$.

We are now ready to prove our main theorem characterizing the behavior of a string-matching automaton on a given input text. As noted above, this theorem shows that the automaton is merely keeping track, at each step, of the longest prefix of the pattern that is a suffix of what has been read so far. In other words, the automaton maintains the invariant (32.5).

Theorem 32.4

If ϕ is the final-state function of a string-matching automaton for a given pattern *P* and $T[1 \dots n]$ is an input text for the automaton, then

 $\phi(T_i) = \sigma(T_i)$

for i = 0, 1, ..., n.

Proof The proof is by induction on *i*. For i = 0, the theorem is trivially true, since $T_0 = \varepsilon$. Thus, $\phi(T_0) = 0 = \sigma(T_0)$.

Now, we assume that $\phi(T_i) = \sigma(T_i)$ and prove that $\phi(T_{i+1}) = \sigma(T_{i+1})$. Let q denote $\phi(T_i)$, and let a denote T[i + 1]. Then,

$$\phi(T_{i+1}) = \phi(T_i a)$$
 (by the definitions of T_{i+1} and a)

$$= \delta(\phi(T_i), a)$$
 (by the definition of ϕ)

$$= \delta(q, a)$$
 (by the definition of q)

$$= \sigma(P_q a)$$
 (by the definition (32.4) of δ)

$$= \sigma(T_i a)$$
 (by Lemma 32.3 and induction)

$$= \sigma(T_{i+1})$$
 (by the definition of T_{i+1}).

By Theorem 32.4, if the machine enters state q on line 4, then q is the largest value such that $P_q \supseteq T_i$. Thus, we have q = m on line 5 if and only if the machine has just scanned an occurrence of the pattern P. We conclude that FINITE-AUTOMATON-MATCHER operates correctly.

Computing the transition function

The following procedure computes the transition function δ from a given pattern P[1..m].

COMPUTE-TRANSITION-FUNCTION (P, Σ)

```
m = P.length
1
2
   for q = 0 to m
3
         for each character a \in \Sigma
4
              k = \min(m + 1, q + 2)
5
              repeat
6
                   k = k - 1
7
              until P_k \supseteq P_q a
8
              \delta(q,a) = k
9
    return \delta
```

This procedure computes $\delta(q, a)$ in a straightforward manner according to its definition in equation (32.4). The nested loops beginning on lines 2 and 3 consider all states q and all characters a, and lines 4–8 set $\delta(q, a)$ to be the largest k such that $P_k \supseteq P_q a$. The code starts with the largest conceivable value of k, which is $\min(m, q + 1)$. It then decreases k until $P_k \supseteq P_q a$, which must eventually occur, since $P_0 = \varepsilon$ is a suffix of every string.

The running time of COMPUTE-TRANSITION-FUNCTION is $O(m^3 |\Sigma|)$, because the outer loops contribute a factor of $m |\Sigma|$, the inner **repeat** loop can run at most m + 1 times, and the test $P_k \supseteq P_q a$ on line 7 can require comparing up

to *m* characters. Much faster procedures exist; by utilizing some cleverly computed information about the pattern *P* (see Exercise 32.4-8), we can improve the time required to compute δ from *P* to $O(m |\Sigma|)$. With this improved procedure for computing δ , we can find all occurrences of a length-*m* pattern in a length-*n* text over an alphabet Σ with $O(m |\Sigma|)$ preprocessing time and $\Theta(n)$ matching time.

Exercises

32.3-1

Construct the string-matching automaton for the pattern P = aabab and illustrate its operation on the text string T = aaababaabaabaabaabaaba.

32.3-2

32.3-3

We call a pattern *P* nonoverlappable if $P_k \supseteq P_q$ implies k = 0 or k = q. Describe the state-transition diagram of the string-matching automaton for a nonoverlappable pattern.

32.3-4 *

Given two patterns P and P', describe how to construct a finite automaton that determines all occurrences of *either* pattern. Try to minimize the number of states in your automaton.

32.3-5

Given a pattern P containing gap characters (see Exercise 32.1-4), show how to build a finite automaton that can find an occurrence of P in a text T in O(n) matching time, where n = |T|.

★ 32.4 The Knuth-Morris-Pratt algorithm

We now present a linear-time string-matching algorithm due to Knuth, Morris, and Pratt. This algorithm avoids computing the transition function δ altogether, and its matching time is $\Theta(n)$ using just an auxiliary function π , which we precompute from the pattern in time $\Theta(m)$ and store in an array $\pi[1 \dots m]$. The array π allows us to compute the transition function δ efficiently (in an amortized sense) "on the fly" as needed. Loosely speaking, for any state $q = 0, 1, \dots, m$ and any character

 $a \in \Sigma$, the value $\pi[q]$ contains the information we need to compute $\delta(q, a)$ but that does not depend on a. Since the array π has only m entries, whereas δ has $\Theta(m |\Sigma|)$ entries, we save a factor of $|\Sigma|$ in the preprocessing time by computing π rather than δ .

The prefix function for a pattern

The prefix function π for a pattern encapsulates knowledge about how the pattern matches against shifts of itself. We can take advantage of this information to avoid testing useless shifts in the naive pattern-matching algorithm and to avoid precomputing the full transition function δ for a string-matching automaton.

Consider the operation of the naive string matcher. Figure 32.10(a) shows a particular shift s of a template containing the pattern P = ababaca against a text T. For this example, q = 5 of the characters have matched successfully, but the 6th pattern character fails to match the corresponding text character. The information that q characters have matched successfully determines the corresponding text characters. Knowing these q text characters allows us to determine immediately that certain shifts are invalid. In the example of the figure, the shift s + 1 is necessarily invalid, since the first pattern character (a) would be aligned with a text character that we know does not match the first pattern character, but does match the second pattern character (b). The shift s' = s + 2 shown in part (b) of the figure, however, aligns the first three pattern characters with three text characters that must necessarily match. In general, it is useful to know the answer to the following question:

Given that pattern characters P[1..q] match text characters T[s+1..s+q], what is the least shift s' > s such that for some k < q,

$$P[1..k] = T[s' + 1..s' + k],$$
(32.6)
where $s' + k = s + q$?

In other words, knowing that $P_q \square T_{s+q}$, we want the longest proper prefix P_k of P_q that is also a suffix of T_{s+q} . (Since s' + k = s + q, if we are given s and q, then finding the smallest shift s' is tantamount to finding the longest prefix length k.) We add the difference q - k in the lengths of these prefixes of P to the shift s to arrive at our new shift s', so that s' = s + (q - k). In the best case, k = 0, so that s' = s + q, and we immediately rule out shifts $s + 1, s + 2, \ldots, s + q - 1$. In any case, at the new shift s' we don't need to compare the first k characters of P with the corresponding characters of T, since equation (32.6) guarantees that they match.

We can precompute the necessary information by comparing the pattern against itself, as Figure 32.10(c) demonstrates. Since $T[s' + 1 \dots s' + k]$ is part of the



Figure 32.10 The prefix function π . (a) The pattern P = ababaca aligns with a text T so that the first q = 5 characters match. Matching characters, shown shaded, are connected by vertical lines. (b) Using only our knowledge of the 5 matched characters, we can deduce that a shift of s + 1 is invalid, but that a shift of s' = s+2 is consistent with everything we know about the text and therefore is potentially valid. (c) We can precompute useful information for such deductions by comparing the pattern with itself. Here, we see that the longest prefix of P that is also a proper suffix of P_5 is P_3 . We represent this precomputed information in the array π , so that $\pi[5] = 3$. Given that q characters have matched successfully at shift s, the next potentially valid shift is at $s' = s + (q - \pi[q])$ as shown in part (b).

known portion of the text, it is a suffix of the string P_q . Therefore, we can interpret equation (32.6) as asking for the greatest k < q such that $P_k \square P_q$. Then, the new shift s' = s + (q-k) is the next potentially valid shift. We will find it convenient to store, for each value of q, the number k of matching characters at the new shift s', rather than storing, say, s' - s.

We formalize the information that we precompute as follows. Given a pattern P[1..m], the *prefix function* for the pattern P is the function $\pi : \{1, 2, ..., m\} \rightarrow \{0, 1, ..., m-1\}$ such that

$$\pi[q] = \max\left\{k : k < q \text{ and } P_k \sqsupset P_q\right\}.$$

That is, $\pi[q]$ is the length of the longest prefix of P that is a proper suffix of P_q . Figure 32.11(a) gives the complete prefix function π for the pattern ababaca.



Figure 32.11 An illustration of Lemma 32.5 for the pattern P = ababaca and q = 5. (a) The π function for the given pattern. Since $\pi[5] = 3$, $\pi[3] = 1$, and $\pi[1] = 0$, by iterating π we obtain $\pi^*[5] = \{3, 1, 0\}$. (b) We slide the template containing the pattern P to the right and note when some prefix P_k of P matches up with some proper suffix of P_5 ; we get matches when k = 3, 1, and 0. In the figure, the first row gives P, and the dotted vertical line is drawn just after P_5 . Successfully matched characters are shown shaded. Vertical lines connect aligned matching characters. Thus, $\{k : k < 5 \text{ and } P_k \square P_5\} = \{3, 1, 0\}$. Lemma 32.5 claims that $\pi^*[q] = \{k : k < q \text{ and } P_k \square P_q\}$ for all q.

The pseudocode below gives the Knuth-Morris-Pratt matching algorithm as the procedure KMP-MATCHER. For the most part, the procedure follows from FINITE-AUTOMATON-MATCHER, as we shall see. KMP-MATCHER calls the auxiliary procedure COMPUTE-PREFIX-FUNCTION to compute π .

KMP-MATCHER(T, P)

1	n = T.length	
2	m = P.length	
3	$\pi = \text{COMPUTE-PREFIX-FUNCTION}(P)$	
4	q = 0	// number of characters matched
5	for $i = 1$ to n	H scan the text from left to right
6	while $q > 0$ and $P[q+1] \neq T[i]$	
7	$q = \pi[q]$	// next character does not match
8	if $P[q+1] == T[i]$	
9	q = q + 1	// next character matches
10	$\mathbf{if} \ q == m$	H is all of P matched?
11	print "Pattern occurs with shift" i	-m
12	$q = \pi[q]$	// look for the next match

COMPUTE-PREFIX-FUNCTION(P)

1 m = P.length2 let $\pi[1..m]$ be a new array 3 $\pi[1] = 0$ $4 \ k = 0$ 5 **for** q = 2 **to** m6 while k > 0 and $P[k + 1] \neq P[q]$ 7 $k = \pi[k]$ **if** P[k + 1] == P[q]8 9 k = k + 1 $\pi[q] = k$ 10 11 return π

These two procedures have much in common, because both match a string against the pattern P: KMP-MATCHER matches the text T against P, and COMPUTE-PREFIX-FUNCTION matches P against itself.

We begin with an analysis of the running times of these procedures. Proving these procedures correct will be more complicated.

Running-time analysis

The running time of COMPUTE-PREFIX-FUNCTION is $\Theta(m)$, which we show by using the aggregate method of amortized analysis (see Section 17.1). The only tricky part is showing that the **while** loop of lines 6–7 executes O(m) times altogether. We shall show that it makes at most m - 1 iterations. We start by making some observations about k. First, line 4 starts k at 0, and the only way that kincreases is by the increment operation in line 9, which executes at most once per iteration of the **for** loop of lines 5–10. Thus, the total increase in k is at most m-1. Second, since k < q upon entering the **for** loop and each iteration of the loop increments q, we always have k < q. Therefore, the assignments in lines 3 and 10 ensure that $\pi[q] < q$ for all $q = 1, 2, \ldots, m$, which means that each iteration of the **while** loop decreases k. Third, k never becomes negative. Putting these facts together, we see that the total decrease in k from the **while** loop is bounded from above by the total increase in k over all iterations of the **for** loop, which is m - 1. Thus, the **while** loop iterates at most m - 1 times in all, and COMPUTE-PREFIX-FUNCTION runs in time $\Theta(m)$.

Exercise 32.4-4 asks you to show, by a similar aggregate analysis, that the matching time of KMP-MATCHER is $\Theta(n)$.

Compared with FINITE-AUTOMATON-MATCHER, by using π rather than δ , we have reduced the time for preprocessing the pattern from $O(m |\Sigma|)$ to $\Theta(m)$, while keeping the actual matching time bounded by $\Theta(n)$.

Correctness of the prefix-function computation

We shall see a little later that the prefix function π helps us simulate the transition function δ in a string-matching automaton. But first, we need to prove that the procedure COMPUTE-PREFIX-FUNCTION does indeed compute the prefix function correctly. In order to do so, we will need to find all prefixes P_k that are proper suffixes of a given prefix P_q . The value of $\pi[q]$ gives us the longest such prefix, but the following lemma, illustrated in Figure 32.11, shows that by iterating the prefix function π , we can indeed enumerate all the prefixes P_k that are proper suffixes of P_q . Let

$$\pi^*[q] = \{\pi[q], \pi^{(2)}[q], \pi^{(3)}[q], \dots, \pi^{(t)}[q]\}$$

where $\pi^{(i)}[q]$ is defined in terms of functional iteration, so that $\pi^{(0)}[q] = q$ and $\pi^{(i)}[q] = \pi[\pi^{(i-1)}[q]]$ for $i \ge 1$, and where the sequence in $\pi^*[q]$ stops upon reaching $\pi^{(i)}[q] = 0$.

Lemma 32.5 (Prefix-function iteration lemma)

Let *P* be a pattern of length *m* with prefix function π . Then, for q = 1, 2, ..., m, we have $\pi^*[q] = \{k : k < q \text{ and } P_k \supseteq P_q\}$.

Proof We first prove that $\pi^*[q] \subseteq \{k : k < q \text{ and } P_k \supseteq P_q\}$ or, equivalently,

$$i \in \pi^*[q] \text{ implies } P_i \sqsupset P_q$$
. (32.7)

If $i \in \pi^*[q]$, then $i = \pi^{(u)}[q]$ for some u > 0. We prove equation (32.7) by induction on u. For u = 1, we have $i = \pi[q]$, and the claim follows since i < qand $P_{\pi[q]} \supseteq P_q$ by the definition of π . Using the relations $\pi[i] < i$ and $P_{\pi[i]} \supseteq P_i$ and the transitivity of < and \supseteq establishes the claim for all i in $\pi^*[q]$. Therefore, $\pi^*[q] \subseteq \{k : k < q \text{ and } P_k \supseteq P_q\}$.

We now prove that $\{k : k < q \text{ and } P_k \square P_q\} \subseteq \pi^*[q]$ by contradiction. Suppose to the contrary that the set $\{k : k < q \text{ and } P_k \square P_q\} - \pi^*[q]$ is nonempty, and let j be the largest number in the set. Because $\pi[q]$ is the largest value in $\{k : k < q \text{ and } P_k \square P_q\}$ and $\pi[q] \in \pi^*[q]$, we must have $j < \pi[q]$, and so we let j' denote the smallest integer in $\pi^*[q]$ that is greater than j. (We can choose $j' = \pi[q]$ if no other number in $\pi^*[q]$ is greater than j.) We have $P_j \square P_q$ because $j \in \{k : k < q \text{ and } P_k \square P_q\}$, and from $j' \in \pi^*[q]$ and equation (32.7), we have $P_{j'} \square P_q$. Thus, $P_j \square P_{j'}$ by Lemma 32.1, and j is the largest value less than j' with this property. Therefore, we must have $\pi[j'] = j$ and, since $j' \in \pi^*[q]$, we must have $j \in \pi^*[q]$ as well. This contradiction proves the lemma.

The algorithm COMPUTE-PREFIX-FUNCTION computes $\pi[q]$, in order, for q = 1, 2, ..., m. Setting $\pi[1]$ to 0 in line 3 of COMPUTE-PREFIX-FUNCTION is certainly correct, since $\pi[q] < q$ for all q. We shall use the following lemma and

its corollary to prove that COMPUTE-PREFIX-FUNCTION computes $\pi[q]$ correctly for q > 1.

Lemma 32.6

Let *P* be a pattern of length *m*, and let π be the prefix function for *P*. For q = 1, 2, ..., m, if $\pi[q] > 0$, then $\pi[q] - 1 \in \pi^*[q - 1]$.

Proof Let $r = \pi[q] > 0$, so that r < q and $P_r \supseteq P_q$; thus, r - 1 < q - 1 and $P_{r-1} \supseteq P_{q-1}$ (by dropping the last character from P_r and P_q , which we can do because r > 0). By Lemma 32.5, therefore, $r - 1 \in \pi^*[q - 1]$. Thus, we have $\pi[q] - 1 = r - 1 \in \pi^*[q - 1]$.

For
$$q = 2, 3, ..., m$$
, define the subset $E_{q-1} \subseteq \pi^*[q-1]$ by
 $E_{q-1} = \{k \in \pi^*[q-1] : P[k+1] = P[q]\}$
 $= \{k : k < q-1 \text{ and } P_k \sqsupset P_{q-1} \text{ and } P[k+1] = P[q]\}$ (by Lemma 32.5)
 $= \{k : k < q-1 \text{ and } P_{k+1} \sqsupset P_q\}$.

The set E_{q-1} consists of the values k < q-1 for which $P_k \supseteq P_{q-1}$ and for which, because P[k + 1] = P[q], we have $P_{k+1} \supseteq P_q$. Thus, E_{q-1} consists of those values $k \in \pi^*[q-1]$ such that we can extend P_k to P_{k+1} and get a proper suffix of P_q .

Corollary 32.7

Let P be a pattern of length m, and let π be the prefix function for P. For $q = 2, 3, \ldots, m$,

$$\pi[q] = \begin{cases} 0 & \text{if } E_{q-1} = \emptyset, \\ 1 + \max\{k \in E_{q-1}\} & \text{if } E_{q-1} \neq \emptyset. \end{cases}$$

Proof If E_{q-1} is empty, there is no $k \in \pi^*[q-1]$ (including k = 0) for which we can extend P_k to P_{k+1} and get a proper suffix of P_q . Therefore $\pi[q] = 0$.

If E_{q-1} is nonempty, then for each $k \in E_{q-1}$ we have k+1 < q and $P_{k+1} \supseteq P_q$. Therefore, from the definition of $\pi[q]$, we have

$$\pi[q] \ge 1 + \max\left\{k \in E_{q-1}\right\} \,. \tag{32.8}$$

Note that $\pi[q] > 0$. Let $r = \pi[q] - 1$, so that $r + 1 = \pi[q]$ and therefore $P_{r+1} \supseteq P_q$. Since r + 1 > 0, we have P[r + 1] = P[q]. Furthermore, by Lemma 32.6, we have $r \in \pi^*[q - 1]$. Therefore, $r \in E_{q-1}$, and so $r \leq \max\{k \in E_{q-1}\}$ or, equivalently,

$$\pi[q] \le 1 + \max\left\{k \in E_{q-1}\right\} \,. \tag{32.9}$$

Combining equations (32.8) and (32.9) completes the proof.

We now finish the proof that COMPUTE-PREFIX-FUNCTION computes π correctly. In the procedure COMPUTE-PREFIX-FUNCTION, at the start of each iteration of the **for** loop of lines 5–10, we have that $k = \pi[q - 1]$. This condition is enforced by lines 3 and 4 when the loop is first entered, and it remains true in each successive iteration because of line 10. Lines 6–9 adjust k so that it becomes the correct value of $\pi[q]$. The **while** loop of lines 6–7 searches through all values $k \in \pi^*[q - 1]$ until it finds a value of k for which P[k + 1] = P[q]; at that point, k is the largest value in the set E_{q-1} , so that, by Corollary 32.7, we can set $\pi[q]$ to k + 1. If the **while** loop cannot find a $k \in \pi^*[q - 1]$ such that P[k + 1] = P[q], then k equals 0 at line 8. If P[1] = P[q], then we should set both k and $\pi[q]$ to 1; otherwise we should leave k alone and set $\pi[q]$ to 0. Lines 8–10 set k and $\pi[q]$ correctly in either case. This completes our proof of the correctness of COMPUTE-PREFIX-FUNCTION.

Correctness of the Knuth-Morris-Pratt algorithm

We can think of the procedure KMP-MATCHER as a reimplemented version of the procedure FINITE-AUTOMATON-MATCHER, but using the prefix function π to compute state transitions. Specifically, we shall prove that in the *i*th iteration of the **for** loops of both KMP-MATCHER and FINITE-AUTOMATON-MATCHER, the state *q* has the same value when we test for equality with *m* (at line 10 in KMP-MATCHER and at line 5 in FINITE-AUTOMATON-MATCHER). Once we have argued that KMP-MATCHER simulates the behavior of FINITE-AUTOMATON-MATCHER, the correctness of KMP-MATCHER follows from the correctness of FINITE-AUTOMATON-MATCHER (though we shall see a little later why line 12 in KMP-MATCHER is necessary).

Before we formally prove that KMP-MATCHER correctly simulates FINITE-AUTOMATON-MATCHER, let's take a moment to understand how the prefix function π replaces the δ transition function. Recall that when a string-matching automaton is in state q and it scans a character a = T[i], it moves to a new state $\delta(q, a)$. If a = P[q + 1], so that a continues to match the pattern, then $\delta(q, a) = q + 1$. Otherwise, $a \neq P[q + 1]$, so that a does not continue to match the pattern, and $0 \leq \delta(q, a) \leq q$. In the first case, when a continues to match, KMP-MATCHER moves to state q + 1 without referring to the π function: the **while** loop test in line 6 comes up false the first time, the test in line 8 comes up true, and line 9 increments q.

The π function comes into play when the character *a* does not continue to match the pattern, so that the new state $\delta(q, a)$ is either *q* or to the left of *q* along the spine of the automaton. The **while** loop of lines 6–7 in KMP-MATCHER iterates through the states in $\pi^*[q]$, stopping either when it arrives in a state, say *q'*, such that *a* matches P[q' + 1] or *q'* has gone all the way down to 0. If *a* matches P[q' + 1],

then line 9 sets the new state to q'+1, which should equal $\delta(q, a)$ for the simulation to work correctly. In other words, the new state $\delta(q, a)$ should be either state 0 or one greater than some state in $\pi^*[q]$.

Let's look at the example in Figures 32.7 and 32.11, which are for the pattern P = ababaca. Suppose that the automaton is in state q = 5; the states in $\pi^*[5]$ are, in descending order, 3, 1, and 0. If the next character scanned is c, then we can easily see that the automaton moves to state $\delta(5, \mathbf{c}) = 6$ in both FINITE-AUTOMATON-MATCHER and KMP-MATCHER. Now suppose that the next character scanned is instead b, so that the automaton should move to state $\delta(5, b) = 4$. The while loop in KMP-MATCHER exits having executed line 7 once, and it arrives in state $q' = \pi[5] = 3$. Since P[q' + 1] = P[4] = b, the test in line 8 comes up true, and KMP-MATCHER moves to the new state $q' + 1 = 4 = \delta(5, b)$. Finally, suppose that the next character scanned is instead a, so that the automaton should move to state $\delta(5, \mathbf{a}) = 1$. The first three times that the test in line 6 executes, the test comes up true. The first time, we find that $P[6] = c \neq a$, and KMP-MATCHER moves to state $\pi[5] = 3$ (the first state in $\pi^*[5]$). The second time, we find that $P[4] = b \neq a$ and move to state $\pi[3] = 1$ (the second state in $\pi^*[5]$). The third time, we find that $P[2] = b \neq a$ and move to state $\pi[1] = 0$ (the last state in $\pi^*[5]$). The **while** loop exits once it arrives in state q' = 0. Now, line 8 finds that P[q'+1] = P[1] = a, and line 9 moves the automaton to the new state $q' + 1 = 1 = \delta(5, a)$.

Thus, our intuition is that KMP-MATCHER iterates through the states in $\pi^*[q]$ in decreasing order, stopping at some state q' and then possibly moving to state q' + 1. Although that might seem like a lot of work just to simulate computing $\delta(q, a)$, bear in mind that asymptotically, KMP-MATCHER is no slower than FINITE-AUTOMATON-MATCHER.

We are now ready to formally prove the correctness of the Knuth-Morris-Pratt algorithm. By Theorem 32.4, we have that $q = \sigma(T_i)$ after each time we execute line 4 of FINITE-AUTOMATON-MATCHER. Therefore, it suffices to show that the same property holds with regard to the **for** loop in KMP-MATCHER. The proof proceeds by induction on the number of loop iterations. Initially, both procedures set q to 0 as they enter their respective **for** loops for the first time. Consider iteration i of the **for** loop in KMP-MATCHER, and let q' be state at the start of this loop iteration. By the inductive hypothesis, we have $q' = \sigma(T_{i-1})$. We need to show that $q = \sigma(T_i)$ at line 10. (Again, we shall handle line 12 separately.)

When we consider the character T[i], the longest prefix of P that is a suffix of T_i is either $P_{q'+1}$ (if P[q'+1] = T[i]) or some prefix (not necessarily proper, and possibly empty) of $P_{q'}$. We consider separately the three cases in which $\sigma(T_i) = 0$, $\sigma(T_i) = q' + 1$, and $0 < \sigma(T_i) \le q'$.

- If $\sigma(T_i) = 0$, then $P_0 = \varepsilon$ is the only prefix of P that is a suffix of T_i . The while loop of lines 6–7 iterates through the values in $\pi^*[q']$, but although $P_q \supseteq T_i$ for every $q \in \pi^*[q']$, the loop never finds a q such that P[q+1] = T[i]. The loop terminates when q reaches 0, and of course line 9 does not execute. Therefore, q = 0 at line 10, so that $q = \sigma(T_i)$.
- If σ(T_i) = q' + 1, then P[q' + 1] = T[i], and the while loop test in line 6 fails the first time through. Line 9 executes, incrementing q so that afterward we have q = q' + 1 = σ(T_i).
- If 0 < σ(T_i) ≤ q', then the while loop of lines 6–7 iterates at least once, checking in decreasing order each value q ∈ π*[q'] until it stops at some q < q'. Thus, P_q is the longest prefix of P_{q'} for which P[q+1] = T[i], so that when the while loop terminates, q + 1 = σ(P_{q'}T[i]). Since q' = σ(T_{i-1}), Lemma 32.3 implies that σ(T_{i-1}T[i]) = σ(P_{q'}T[i]). Thus, we have

$$q + 1 = \sigma(P_{q'}T[i])$$

= $\sigma(T_{i-1}T[i])$
= $\sigma(T_i)$

when the **while** loop terminates. After line 9 increments q, we have $q = \sigma(T_i)$.

Line 12 is necessary in KMP-MATCHER, because otherwise, we might reference P[m + 1] on line 6 after finding an occurrence of P. (The argument that $q = \sigma(T_{i-1})$ upon the next execution of line 6 remains valid by the hint given in Exercise 32.4-8: $\delta(m, a) = \delta(\pi[m], a)$ or, equivalently, $\sigma(Pa) = \sigma(P_{\pi[m]}a)$ for any $a \in \Sigma$.) The remaining argument for the correctness of the Knuth-Morris-Pratt algorithm follows from the correctness of FINITE-AUTOMATON-MATCHER, since we have shown that KMP-MATCHER simulates the behavior of FINITE-AUTOMATON-MATCHER.

Exercises

32.4-1

32.4-2

Give an upper bound on the size of $\pi^*[q]$ as a function of q. Give an example to show that your bound is tight.

32.4-3

Explain how to determine the occurrences of pattern P in the text T by examining the π function for the string PT (the string of length m+n that is the concatenation of P and T).

32.4-4

Use an aggregate analysis to show that the running time of KMP-MATCHER is $\Theta(n)$.

32.4-5

Use a potential function to show that the running time of KMP-MATCHER is $\Theta(n)$.

32.4-6

Show how to improve KMP-MATCHER by replacing the occurrence of π in line 7 (but not line 12) by π' , where π' is defined recursively for q = 1, 2, ..., m - 1 by the equation

$$\pi'[q] = \begin{cases} 0 & \text{if } \pi[q] = 0, \\ \pi'[\pi[q]] & \text{if } \pi[q] \neq 0 \text{ and } P[\pi[q] + 1] = P[q + 1], \\ \pi[q] & \text{if } \pi[q] \neq 0 \text{ and } P[\pi[q] + 1] \neq P[q + 1]. \end{cases}$$

Explain why the modified algorithm is correct, and explain in what sense this change constitutes an improvement.

32.4-7

Give a linear-time algorithm to determine whether a text T is a cyclic rotation of another string T'. For example, arc and car are cyclic rotations of each other.

32.4-8 *

Give an $O(m |\Sigma|)$ -time algorithm for computing the transition function δ for the string-matching automaton corresponding to a given pattern *P*. (*Hint:* Prove that $\delta(q, a) = \delta(\pi[q], a)$ if q = m or $P[q + 1] \neq a$.)

Problems

32-1 String matching based on repetition factors

Let y^i denote the concatenation of string y with itself i times. For example, (ab)³ = ababab. We say that a string $x \in \Sigma^*$ has *repetition factor* r if $x = y^r$ for some string $y \in \Sigma^*$ and some r > 0. Let $\rho(x)$ denote the largest r such that x has repetition factor r.

a. Give an efficient algorithm that takes as input a pattern P[1..m] and computes the value $\rho(P_i)$ for i = 1, 2, ..., m. What is the running time of your algorithm?

- **b.** For any pattern P[1 ...m], let $\rho^*(P)$ be defined as $\max_{1 \le i \le m} \rho(P_i)$. Prove that if the pattern P is chosen randomly from the set of all binary strings of length m, then the expected value of $\rho^*(P)$ is O(1).
- c. Argue that the following string-matching algorithm correctly finds all occurrences of pattern P in a text T[1..n] in time $O(\rho^*(P)n + m)$:

REPETITION-MATCHER (P, T)

1 m = P.length $2 \quad n = T.length$ 3 $k = 1 + \rho^*(P)$ 4 q = 05 s = 06 while s < n - m7 **if** T[s+q+1] == P[q+1]8 q = q + 19 if q == m10 print "Pattern occurs with shift" s 11 if q == m or $T[s + q + 1] \neq P[q + 1]$ 12 $s = s + \max(1, \lceil q/k \rceil)$ 13 q = 0

This algorithm is due to Galil and Seiferas. By extending these ideas greatly, they obtained a linear-time string-matching algorithm that uses only O(1) storage beyond what is required for P and T.

Chapter notes

The relation of string matching to the theory of finite automata is discussed by Aho, Hopcroft, and Ullman [5]. The Knuth-Morris-Pratt algorithm [214] was invented independently by Knuth and Pratt and by Morris; they published their work jointly. Reingold, Urban, and Gries [294] give an alternative treatment of the Knuth-Morris-Pratt algorithm. The Rabin-Karp algorithm was proposed by Karp and Rabin [201]. Galil and Seiferas [126] give an interesting deterministic linear-time string-matching algorithm that uses only O(1) space beyond that required to store the pattern and text.