

1

Boolean retrieval

The meaning of the term *information retrieval* can be very broad. Just getting a credit card out of your wallet so that you can type in the card number is a form of information retrieval. However, as an academic field of study, *information retrieval* might be defined thus:

Information retrieval (IR) is finding material (usually documents) of an unstructured nature (usually text) that satisfies an information need from within large collections (usually stored on computers).

As defined in this way, information retrieval used to be an activity that only a few people engaged in: reference librarians, paralegals, and similar professional searchers. Now the world has changed, and hundreds of millions of people engage in information retrieval every day when they use a web search engine or search their email.¹ Information retrieval is fast becoming the dominant form of information access, overtaking traditional database-style searching (the sort that is going on when a clerk says to you: “I’m sorry, I can only look up your order if you can give me your Order ID”).

IR can also cover other kinds of data and information problems beyond that specified in the core definition above. The term “unstructured data” refers to data which does not have clear, semantically overt, easy-for-a-computer structure. It is the opposite of structured data, the canonical example of which is a relational database, of the sort companies usually use to maintain product inventories and personnel records. In reality, almost no data are truly “unstructured”. This is definitely true of all text data if you count the latent linguistic structure of human languages. But even accepting that the intended notion of structure is overt structure, most text has structure, such as headings and paragraphs and footnotes, which is commonly represented in documents by explicit markup (such as the coding underlying web

1. In modern parlance, the word “search” has tended to replace “(information) retrieval”; the term “search” is quite ambiguous, but in context we use the two synonymously.

pages). IR is also used to facilitate “semistructured” search such as finding a document where the title contains Java and the body contains threading.

The field of information retrieval also covers supporting users in browsing or filtering document collections or further processing a set of retrieved documents. Given a set of documents, clustering is the task of coming up with a good grouping of the documents based on their contents. It is similar to arranging books on a bookshelf according to their topic. Given a set of topics, standing information needs, or other categories (such as suitability of texts for different age groups), classification is the task of deciding which class(es), if any, each of a set of documents belongs to. It is often approached by first manually classifying some documents and then hoping to be able to classify new documents automatically.

Information retrieval systems can also be distinguished by the scale at which they operate, and it is useful to distinguish three prominent scales. In *web search*, the system has to provide search over billions of documents stored on millions of computers. Distinctive issues are needing to gather documents for indexing, being able to build systems that work efficiently at this enormous scale, and handling particular aspects of the web, such as the exploitation of hypertext and not being fooled by site providers manipulating page content in an attempt to boost their search engine rankings, given the commercial importance of the web. We focus on all these issues in Chapters 19–21. At the other extreme is *personal information retrieval*. In the last few years, consumer operating systems have integrated information retrieval (such as Apple’s Mac OS X Spotlight or Windows Vista’s Instant Search). Email programs usually not only provide search but also text classification: they at least provide a spam (junk mail) filter, and commonly also provide either manual or automatic means for classifying mail so that it can be placed directly into particular folders. Distinctive issues here include handling the broad range of document types on a typical personal computer, and making the search system maintenance free and sufficiently lightweight in terms of startup, processing, and disk space usage that it can run on one machine without annoying its owner. In between is the space of *enterprise, institutional, and domain-specific search*, where retrieval might be provided for collections such as a corporation’s internal documents, a database of patents, or research articles on biochemistry. In this case, the documents will typically be stored on centralized file systems and one or a handful of dedicated machines will provide search over the collection. This book contains techniques of value over this whole spectrum, but our coverage of some aspects of parallel and distributed search in web-scale search systems is comparatively light owing to the relatively small published literature on the details of such systems. However, outside of a handful of web search companies, a software developer is most likely to encounter the personal search and enterprise scenarios.

In this chapter we begin with a very simple example of an information retrieval problem, and introduce the idea of a term-document matrix (Section 1.1) and the central inverted index data structure (Section 1.2). We will then examine the Boolean retrieval model and how Boolean queries are processed (Sections 1.3 and 1.4).

1.1 An example information retrieval problem

A fat book which many people own is Shakespeare’s Collected Works. Suppose you wanted to determine which plays of Shakespeare contain the words Brutus AND Caesar AND NOT Calpurnia. One way to do that is to start at the beginning and to read through all the text, noting for each play whether it contains Brutus and Caesar and excluding it from consideration if it contains Calpurnia. The simplest form of document retrieval is for a computer to do this sort of linear scan through documents. This process is commonly referred to as *grepping* through text, after the Unix command `grep`, which performs this process. Grepping through text can be a very effective process, especially given the speed of modern computers, and often allows useful possibilities for wildcard pattern matching through the use of regular expressions. With modern computers, for simple querying of modest collections (the size of Shakespeare’s Collected Works is a bit under one million words of text in total), you really need nothing more.

But for many purposes, you do need more:

1. To process large document collections quickly. The amount of online data has grown at least as quickly as the speed of computers, and we would now like to be able to search collections that total in the order of billions to trillions of words.
2. To allow more flexible matching operations. For example, it is impractical to perform the query Romans NEAR countrymen with `grep`, where NEAR might be defined as “within 5 words” or “within the same sentence”.
3. To allow ranked retrieval: in many cases you want the best answer to an information need among many documents that contain certain words.

INDEX

The way to avoid linearly scanning the texts for each query is to *index* the documents in advance. Let us stick with Shakespeare’s Collected Works, and use it to introduce the basics of the Boolean retrieval model. Suppose we record for each document – here a play of Shakespeare’s – whether it contains each word out of all the words Shakespeare used (Shakespeare used about 32,000 different words). The result is a binary term-document *incidence matrix*, as in Figure 1.1. *Terms* are the indexed units (further discussed in Section 2.2); they are usually words, and for the moment you can think of

INCIDENCE MATRIX
TERM

	Antony and Cleopatra	Julius Caesar	The Tempest	Hamlet	Othello	Macbeth	...
Antony	1	1	0	0	0	1	
Brutus	1	1	0	1	0	0	
Caesar	1	1	0	1	1	1	
Calpurnia	0	1	0	0	0	0	
Cleopatra	1	0	0	0	0	0	
mercy	1	0	1	1	1	1	
worser	1	0	1	1	1	0	
...							

► **Figure 1.1** A term-document incidence matrix. Matrix element (t, d) is 1 if the play in column d contains the word in row t , and is 0 otherwise.

them as words, but the information retrieval literature normally speaks of terms because some of them, such as perhaps I-9 or Hong Kong are not usually thought of as words. Now, depending on whether we look at the matrix rows or columns, we can have a vector for each term, which shows the documents it appears in, or a vector for each document, showing the terms that occur in it.²

To answer the query Brutus AND Caesar AND NOT Calpurnia, we take the vectors for Brutus, Caesar and Calpurnia, complement the last, and then do a bitwise AND:

$$110100 \text{ AND } 110111 \text{ AND } 101111 = 100100$$

The answers for this query are thus *Antony and Cleopatra* and *Hamlet* (Figure 1.2).

The *Boolean retrieval model* is a model for information retrieval in which we can pose any query which is in the form of a Boolean expression of terms, that is, in which terms are combined with the operators AND, OR, and NOT. The model views each document as just a set of words.

Let us now consider a more realistic scenario, simultaneously using the opportunity to introduce some terminology and notation. Suppose we have $N = 1$ million documents. By *documents* we mean whatever units we have decided to build a retrieval system over. They might be individual memos or chapters of a book (see Section 2.1.2 (page 20) for further discussion). We will refer to the group of documents over which we perform retrieval as the (document) *collection*. It is sometimes also referred to as a *corpus* (a *body* of texts). Suppose each document is about 1000 words long (2–3 book pages). If

2. Formally, we take the transpose of the matrix to be able to get the terms as column vectors.

Antony and Cleopatra, Act III, Scene ii

Agrippa [Aside to Domitius Enobarbus]: Why, Enobarbus,
 When Antony found Julius Caesar dead,
 He cried almost to roaring; and he wept
 When at Philippi he found Brutus slain.

Hamlet, Act III, Scene ii

Lord Polonius: I did enact Julius Caesar: I was killed i' the
 Capitol; Brutus killed me.

► **Figure 1.2** Results from Shakespeare for the query `Brutus AND Caesar AND NOT Calpurnia`.

we assume an average of 6 bytes per word including spaces and punctuation, then this is a document collection about 6 GB in size. Typically, there might be about $M = 500,000$ distinct terms in these documents. There is nothing special about the numbers we have chosen, and they might vary by an order of magnitude or more, but they give us some idea of the dimensions of the kinds of problems we need to handle. We will discuss and model these size assumptions in Section 5.1 (page 86).

Our goal is to develop a system to address the *ad hoc retrieval* task. This is the most standard IR task. In it, a system aims to provide documents from within the collection that are relevant to an arbitrary user information need, communicated to the system by means of a one-off, user-initiated query. An *information need* is the topic about which the user desires to know more, and is differentiated from a *query*, which is what the user conveys to the computer in an attempt to communicate the information need. A document is *relevant* if it is one that the user perceives as containing information of value with respect to their personal information need. Our example above was rather artificial in that the information need was defined in terms of particular words, whereas usually a user is interested in a topic like “pipeline leaks” and would like to find relevant documents regardless of whether they precisely use those words or express the concept with other words such as pipeline rupture. To assess the *effectiveness* of an IR system (i.e., the quality of its search results), a user will usually want to know two key statistics about the system’s returned results for a query:

AD HOC RETRIEVAL

INFORMATION NEED

QUERY

RELEVANCE

EFFECTIVENESS

Precision

Recall

Precision: What fraction of the returned results are relevant to the information need?

Recall: What fraction of the relevant documents in the collection were returned by the system?

Detailed discussion of relevance and evaluation measures including precision and recall is found in Chapter 8.

We now cannot build a term-document matrix in a naive way. A $500K \times 1M$ matrix has half-a-trillion 0's and 1's – too many to fit in a computer's memory. But the crucial observation is that the matrix is extremely sparse, that is, it has few non-zero entries. Because each document is 1000 words long, the matrix has no more than one billion 1's, so a minimum of 99.8% of the cells are zero. A much better representation is to record only the things that do occur, that is, the 1 positions.

INVERTED INDEX

DICTIONARY VOCABULARY LEXICON

POSTING POSTINGS LIST POSTINGS

This idea is central to the first major concept in information retrieval, the *inverted index*. The name is actually redundant: an index always maps back from terms to the parts of a document where they occur. Nevertheless, *inverted index*, or sometimes *inverted file*, has become the standard term in information retrieval.³ The basic idea of an inverted index is shown in Figure 1.3. We keep a *dictionary* of terms (sometimes also referred to as a *vocabulary* or *lexicon*; in this book, we use *dictionary* for the data structure and *vocabulary* for the set of terms). Then for each term, we have a list that records which documents the term occurs in. Each item in the list – which records that a term appeared in a document (and, later, often, the positions in the document) – is conventionally called a *posting*.⁴ The list is then called a *postings list* (or inverted list), and all the postings lists taken together are referred to as the *postings*. The dictionary in Figure 1.3 has been sorted alphabetically and each postings list is sorted by document ID. We will see why this is useful in Section 1.3, below, but later we will also consider alternatives to doing this (Section 7.1.5).

1.2 A first take at building an inverted index

To gain the speed benefits of indexing at retrieval time, we have to build the index in advance. The major steps in this are:

1. Collect the documents to be indexed:

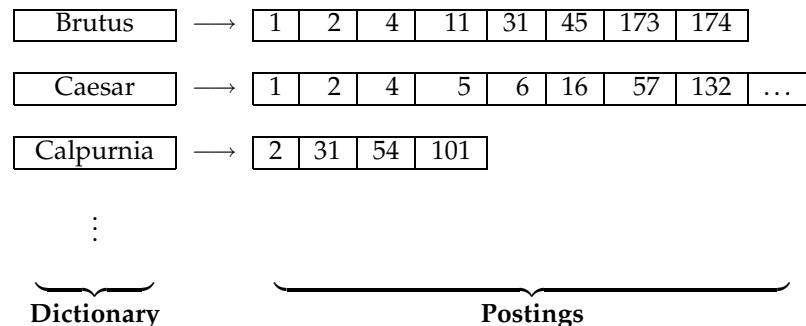
Friends, Romans, countrymen.	So let it be with Caesar	...
------------------------------	--------------------------	-----

2. Tokenize the text, turning each document into a list of tokens:

Friends	Romans	countrymen	So	...
---------	--------	------------	----	-----

3. Some information retrieval researchers prefer the term *inverted file*, but expressions like index construction and index compression are much more common than *inverted file* construction and *inverted file* compression. For consistency, we use *(inverted) index* throughout this book.

4. In a (non-positional) inverted index, a posting is just a document ID, but it is inherently associated with a term, via the postings list it is placed on; sometimes we will also talk of a (term, docID) pair as a posting.



► **Figure 1.3** The two parts of an inverted index. The dictionary is commonly kept in memory, with pointers to each postings list, which is stored on disk.

3. Do linguistic preprocessing, producing a list of normalized tokens, which are the indexing terms: friend roman countryman so ...
4. Index the documents that each term occurs in by creating an inverted index, consisting of a dictionary and postings.

We will define and discuss the earlier stages of processing, that is, steps 1–3, in Section 2.2 (page 22). Until then you can think of *tokens* and *normalized tokens* as also loosely equivalent to *words*. Here, we assume that the first 3 steps have already been done, and we examine building a basic inverted index by sort-based indexing.

Within a document collection, we assume that each document has a unique serial number, known as the document identifier (*docID*). During index construction, we can simply assign successive integers to each new document when it is first encountered. The input to indexing is a list of normalized tokens for each document, which we can equally think of as a list of pairs of term and docID, as in Figure 1.4. The core indexing step is *sorting* this list so that the terms are alphabetical, giving us the representation in the middle column of Figure 1.4. Multiple occurrences of the same term from the same document are then merged.⁵ Instances of the same term are then grouped, and the result is split into a *dictionary* and *postings*, as shown in the right column of Figure 1.4. Since a term generally occurs in a number of documents, this data organization already reduces the storage requirements of the index. The dictionary also records some statistics, such as the number of documents which contain each term (the *document frequency*, which is here also the length of each postings list). This information is not vital for a basic Boolean search engine, but it allows us to improve the efficiency of the

5. Unix users can note that these steps are similar to use of the `sort` and then `uniq` commands.

Doc 1

I did enact Julius Caesar: I was killed
i' the Capitol; Brutus killed me.

term	docID	term	docID
I	1	ambitious	2
did	1	be	2
enact	1	brutus	1
julius	1	brutus	2
caesar	1	capitol	1
I	1	caesar	1
was	1	caesar	2
killed	1	caesar	2
i'	1	did	1
the	1	enact	1
capitol	1	hath	1
brutus	1	I	1
killed	1	I	1
me	1	i'	1
so	2	it	2
let	2	julius	1
it	2	killed	1
be	2	killed	1
with	2	let	2
caesar	2	me	1
the	2	noble	2
noble	2	so	2
brutus	2	the	1
hath	2	the	2
told	2	told	2
you	2	you	2
caesar	2	was	1
was	2	was	2
ambitious	2	with	2

Doc 2

So let it be with Caesar. The noble Brutus
hath told you Caesar was ambitious:

term	doc. freq.	→	postings lists
ambitious	1	→	[2]
be	1	→	[2]
brutus	2	→	[1] → [2]
capitol	1	→	[1]
caesar	2	→	[1] → [2]
did	1	→	[1]
enact	1	→	[1]
hath	1	→	[2]
I	1	→	[1]
i'	1	→	[1]
it	1	→	[2]
julius	1	→	[1]
killed	1	→	[1]
let	1	→	[2]
me	1	→	[1]
noble	1	→	[2]
so	1	→	[2]
the	2	→	[1] → [2]
told	1	→	[2]
you	1	→	[2]
was	2	→	[1] → [2]
with	1	→	[2]

► **Figure 1.4** Building an index by sorting and grouping. The sequence of terms in each document, tagged by their documentID (left) is sorted alphabetically (middle). Instances of the same term are then grouped by word and then by documentID. The terms and documentIDs are then separated out (right). The dictionary stores the terms, and has a pointer to the postings list for each term. It commonly also stores other summary information such as, here, the document frequency of each term. We use this information for improving query time efficiency and, later, for weighting in ranked retrieval models. Each postings list stores the list of documents in which a term occurs, and may store other information such as the term frequency (the frequency of each term in each document) or the position(s) of the term in each document.

search engine at query time, and it is a statistic later used in many ranked retrieval models. The postings are secondarily sorted by docID. This provides the basis for efficient query processing. This inverted index structure is essentially without rivals as the most efficient structure for supporting ad hoc text search.

In the resulting index, we pay for storage of both the dictionary and the postings lists. The latter are much larger, but the dictionary is commonly kept in memory, while postings lists are normally kept on disk, so the size of each is important, and in Chapter 5 we will examine how each can be optimized for storage and access efficiency. What data structure should be used for a postings list? A fixed length array would be wasteful as some words occur in many documents, and others in very few. For an in-memory postings list, two good alternatives are singly linked lists or variable length arrays. Singly linked lists allow cheap insertion of documents into postings lists (following updates, such as when recrawling the web for updated documents), and naturally extend to more advanced indexing strategies such as skip lists (Section 2.3), which require additional pointers. Variable length arrays win in space requirements by avoiding the overhead for pointers and in time requirements because their use of contiguous memory increases speed on modern processors with memory caches. Extra pointers can in practice be encoded into the lists as offsets. If updates are relatively infrequent, variable length arrays will be more compact and faster to traverse. We can also use a hybrid scheme with a linked list of fixed length arrays for each term. When postings lists are stored on disk, they are stored (perhaps compressed) as a contiguous run of postings without explicit pointers (as in Figure 1.3), so as to minimize the size of the postings list and the number of disk seeks to read a postings list into memory.



Exercise 1.1

[\star]

Draw the inverted index that would be built for the following document collection. (See Figure 1.3 for an example.)

Doc 1 new home sales top forecasts
Doc 2 home sales rise in july
Doc 3 increase in home sales in july
Doc 4 july new home sales rise

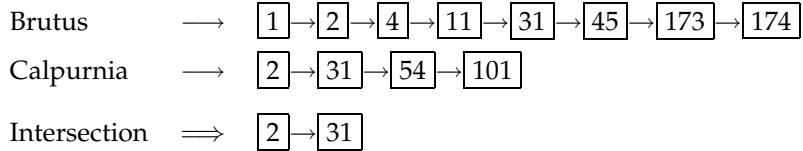
Exercise 1.2

[\star]

Consider these documents:

Doc 1 breakthrough drug for schizophrenia
Doc 2 new schizophrenia drug
Doc 3 new approach for treatment of schizophrenia
Doc 4 new hopes for schizophrenia patients

- Draw the term-document incidence matrix for this document collection.



► **Figure 1.5** Intersecting the postings lists for Brutus and Calpurnia from Figure 1.3.

- b. Draw the inverted index representation for this collection, as in Figure 1.3 (page 7).

Exercise 1.3

[*]

For the document collection shown in Exercise 1.2, what are the returned results for these queries:

- schizophrenia AND drug
- for AND NOT(drug OR approach)

1.3 Processing Boolean queries

SIMPLE CONJUNCTIVE
QUERIES
(1.1)

How do we process a query using an inverted index and the basic Boolean retrieval model? Consider processing the *simple conjunctive query*:

Brutus AND Calpurnia

over the inverted index partially shown in Figure 1.3 (page 7). We:

- Locate Brutus in the Dictionary
- Retrieve its postings
- Locate Calpurnia in the Dictionary
- Retrieve its postings
- Intersect the two postings lists, as shown in Figure 1.5.

POSTINGS LIST
INTERSECTION
POSTINGS MERGE

The *intersection* operation is the crucial one: we need to efficiently intersect postings lists so as to be able to quickly find documents that contain both terms. (This operation is sometimes referred to as *merging* postings lists: this slightly counterintuitive name reflects using the term *merge algorithm* for a general family of algorithms that combine multiple sorted lists by interleaved advancing of pointers through each; here we are merging the lists with a logical AND operation.)

There is a simple and effective method of intersecting postings lists using the merge algorithm (see Figure 1.6): we maintain pointers into both lists

```

INTERSECT( $p_1, p_2$ )
1  $answer \leftarrow \langle \rangle$ 
2 while  $p_1 \neq \text{NIL}$  and  $p_2 \neq \text{NIL}$ 
3 do if  $docID(p_1) = docID(p_2)$ 
4   then ADD( $answer, docID(p_1)$ )
5      $p_1 \leftarrow next(p_1)$ 
6      $p_2 \leftarrow next(p_2)$ 
7   else if  $docID(p_1) < docID(p_2)$ 
8     then  $p_1 \leftarrow next(p_1)$ 
9   else  $p_2 \leftarrow next(p_2)$ 
10 return  $answer$ 

```

► **Figure 1.6** Algorithm for the intersection of two postings lists p_1 and p_2 .

and walk through the two postings lists simultaneously, in time linear in the total number of postings entries. At each step, we compare the docID pointed to by both pointers. If they are the same, we put that docID in the results list, and advance both pointers. Otherwise we advance the pointer pointing to the smaller docID. If the lengths of the postings lists are x and y , the intersection takes $O(x + y)$ operations. Formally, the complexity of querying is $\Theta(N)$, where N is the number of documents in the collection.⁶ Our indexing methods gain us just a constant, not a difference in Θ time complexity compared to a linear scan, but in practice the constant is huge. To use this algorithm, it is crucial that postings be sorted by a single global ordering. Using a numeric sort by docID is one simple way to achieve this.

We can extend the intersection operation to process more complicated queries like:

(1.2) (Brutus OR Caesar) AND NOT Calpurnia

QUERY OPTIMIZATION

Query optimization is the process of selecting how to organize the work of answering a query so that the least total amount of work needs to be done by the system. A major element of this for Boolean queries is the order in which postings lists are accessed. What is the best order for query processing? Consider a query that is an AND of t terms, for instance:

(1.3) Brutus AND Caesar AND Calpurnia

For each of the t terms, we need to get its postings, then AND them together. The standard heuristic is to process terms in order of increasing document

6. The notation $\Theta(\cdot)$ is used to express an asymptotically tight bound on the complexity of an algorithm. Informally, this is often written as $O(\cdot)$, but this notation really expresses an asymptotic upper bound, which need not be tight (Cormen et al. 1990).

```

INTERSECT( $\langle t_1, \dots, t_n \rangle$ )
1  $terms \leftarrow \text{SORTBYINCREASINGFREQUENCY}(\langle t_1, \dots, t_n \rangle)$ 
2  $result \leftarrow \text{postings}(first(terms))$ 
3  $terms \leftarrow rest(terms)$ 
4 while  $terms \neq \text{NIL}$  and  $result \neq \text{NIL}$ 
5   do  $result \leftarrow \text{INTERSECT}(result, \text{postings}(first(terms)))$ 
6    $terms \leftarrow rest(terms)$ 
7 return  $result$ 

```

► **Figure 1.7** Algorithm for conjunctive queries that returns the set of documents containing each term in the input list of terms.

frequency: if we start by intersecting the two smallest postings lists, then all intermediate results must be no bigger than the smallest postings list, and we are therefore likely to do the least amount of total work. So, for the postings lists in Figure 1.3 (page 7), we execute the above query as:

(1.4) (Calpurnia AND Brutus) AND Caesar

This is a first justification for keeping the frequency of terms in the dictionary: it allows us to make this ordering decision based on in-memory data before accessing any postings list.

Consider now the optimization of more general queries, such as:

(1.5) (madding OR crowd) AND (ignoble OR strife) AND (killed OR slain)

As before, we will get the frequencies for all terms, and we can then (conservatively) estimate the size of each OR by the sum of the frequencies of its disjuncts. We can then process the query in increasing order of the size of each disjunctive term.

For arbitrary Boolean queries, we have to evaluate and temporarily store the answers for intermediate expressions in a complex expression. However, in many circumstances, either because of the nature of the query language, or just because this is the most common type of query that users submit, a query is purely conjunctive. In this case, rather than viewing merging postings lists as a function with two inputs and a distinct output, it is more efficient to intersect each retrieved postings list with the current intermediate result in memory, where we initialize the intermediate result by loading the postings list of the least frequent term. This algorithm is shown in Figure 1.7. The intersection operation is then asymmetric: the intermediate results list is in memory while the list it is being intersected with is being read from disk. Moreover the intermediate results list is always at least as short as the other list, and in many cases it is orders of magnitude shorter. The postings

intersection can still be done by the algorithm in Figure 1.6, but when the difference between the list lengths is very large, opportunities to use alternative techniques open up. The intersection can be calculated in place by destructively modifying or marking invalid items in the intermediate results list. Or the intersection can be done as a sequence of binary searches in the long postings lists for each posting in the intermediate results list. Another possibility is to store the long postings list as a hashtable, so that membership of an intermediate result item can be calculated in constant rather than linear or log time. However, such alternative techniques are difficult to combine with postings list compression of the sort discussed in Chapter 5. Moreover, standard postings list intersection operations remain necessary when both terms of a query are very common.


Exercise 1.4
[\star]

For the queries below, can we still run through the intersection in time $O(x + y)$, where x and y are the lengths of the postings lists for Brutus and Caesar? If not, what can we achieve?

- a. Brutus AND NOT Caesar
- b. Brutus OR NOT Caesar

Exercise 1.5
[\star]

Extend the postings merge algorithm to arbitrary Boolean query formulas. What is its time complexity? For instance, consider:

- c. (Brutus OR Caesar) AND NOT (Antony OR Cleopatra)

Can we always merge in linear time? Linear in what? Can we do better than this?

Exercise 1.6
[$\star\star$]

We can use distributive laws for AND and OR to rewrite queries.

- a. Show how to rewrite the query in Exercise 1.5 into disjunctive normal form using the distributive laws.
- b. Would the resulting query be more or less efficiently evaluated than the original form of this query?
- c. Is this result true in general or does it depend on the words and the contents of the document collection?

Exercise 1.7
[\star]

Recommend a query processing order for

- d. (tangerine OR trees) AND (marmalade OR skies) AND (kaleidoscope OR eyes)

given the following postings list sizes:

Term	Postings size
eyes	213312
kaleidoscope	87009
marmalade	107913
skies	271658
tangerine	46653
trees	316812

Exercise 1.8

[★]

If the query is:

- e. friends AND romans AND (NOT countrymen)

how could we use the frequency of countrymen in evaluating the best query evaluation order? In particular, propose a way of handling negation in determining the order of query processing.

Exercise 1.9

[★★]

For a conjunctive query, is processing postings lists in order of size guaranteed to be optimal? Explain why it is, or give an example where it isn't.

Exercise 1.10

[★★]

Write out a postings merge algorithm, in the style of Figure 1.6 (page 11), for an $x \text{ OR } y$ query.

Exercise 1.11

[★★]

How should the Boolean query $x \text{ AND NOT } y$ be handled? Why is naive evaluation of this query normally very expensive? Write out a postings merge algorithm that evaluates this query efficiently.

1.4 The extended Boolean model versus ranked retrieval

RANKED RETRIEVAL
MODEL
FREE TEXT QUERIES

PROXIMITY OPERATOR

The Boolean retrieval model contrasts with *ranked retrieval models* such as the vector space model (Section 6.3), in which users largely use *free text queries*, that is, just typing one or more words rather than using a precise language with operators for building up query expressions, and the system decides which documents best satisfy the query. Despite decades of academic research on the advantages of ranked retrieval, systems implementing the Boolean retrieval model were the main or only search option provided by large commercial information providers for three decades until the early 1990s (approximately the date of arrival of the World Wide Web). However, these systems did not have just the basic Boolean operations (AND, OR, and NOT) which we have presented so far. A strict Boolean expression over terms with an unordered results set is too limited for many of the information needs that people have, and these systems implemented extended Boolean retrieval models by incorporating additional operators such as term proximity operators. A *proximity operator* is a way of specifying that two terms in a query

must occur close to each other in a document, where closeness may be measured by limiting the allowed number of intervening words or by reference to a structural unit such as a sentence or paragraph.

 **Example 1.1: Commercial Boolean searching: Westlaw.** Westlaw (<http://www.westlaw.com/>) is the largest commercial legal search service (in terms of the number of paying subscribers), with over half a million subscribers performing millions of searches a day over tens of terabytes of text data. The service was started in 1975. In 2005, Boolean search (called “Terms and Connectors” by Westlaw) was still the default, and used by a large percentage of users, although ranked free text querying (called “Natural Language” by Westlaw) was added in 1992. Here are some example Boolean queries on Westlaw:

Information need: Information on the legal theories involved in preventing the disclosure of trade secrets by employees formerly employed by a competing company. *Query:* "trade secret" /s disclos! /s prevent /s employe!

Information need: Requirements for disabled people to be able to access a work-place.
Query: disab! /p access! /s work-site work-place (employment /3 place)

Information need: Cases about a host's responsibility for drunk guests.
Query: host! /p (responsib! liab!) /p (intoxicat! drunk!) /p guest

Note the long, precise queries and the use of proximity operators, both uncommon in web search. Submitted queries average about ten words in length. Unlike web search conventions, a space between words represents disjunction (the tightest binding operator), & is AND and /s, /p, and /k ask for matches in the same sentence, same paragraph or within k words respectively. Double quotes give a *phrase search* (consecutive words); see Section 2.4 (page 39). The exclamation mark (!) gives a trailing wildcard query (see Section 3.2, page 51); thus liab! matches all words starting with liab. Additionally work-site matches any of *worksit*e, *work-site* or *work site*; see Section 2.2.1 (page 22). Typical expert queries are usually carefully defined and incrementally developed until they obtain what look to be good results to the user.

Many users, particularly professionals, prefer Boolean query models. Boolean queries are precise: a document either matches the query or it does not. This offers the user greater control and transparency over what is retrieved. And some domains, such as legal materials, allow an effective means of document ranking within a Boolean model: Westlaw returns documents in reverse chronological order, which is in practice quite effective. In 2007, the majority of law librarians still seem to recommend terms and connectors for high recall searches, and the majority of legal users think they are getting greater control by using them. However, this does not mean that Boolean queries are more effective for professional searchers. Indeed, experimenting on a Westlaw subcollection, [Turtle \(1994\)](#) found that free text queries produced better results than Boolean queries prepared by Westlaw's own reference librarians for the majority of the information needs in his experiments. A general problem with Boolean search is that using AND operators tends to produce high precision but low recall searches, while using OR operators gives low precision but high recall searches, and it is difficult or impossible to find a satisfactory middle ground.

In this chapter, we have looked at the structure and construction of a basic

inverted index, comprising a dictionary and postings lists. We introduced the Boolean retrieval model, and examined how to do efficient retrieval via linear time merges and simple query optimization. In Chapters 2–7 we will consider in detail richer query models and the sort of augmented index structures that are needed to handle them efficiently. Here we just mention a few of the main additional things we would like to be able to do:

1. We would like to better determine the set of terms in the dictionary and to provide retrieval that is tolerant to spelling mistakes and inconsistent choice of words.
2. It is often useful to search for compounds or phrases that denote a concept such as “operating system”. As the Westlaw examples show, we might also wish to do proximity queries such as Gates NEAR Microsoft. To answer such queries, the index has to be augmented to capture the proximities of terms in documents.
3. A Boolean model only records term presence or absence, but often we would like to accumulate evidence, giving more weight to documents that have a term several times as opposed to ones that contain it only once. To be able to do this we need *term frequency* information (the number of times a term occurs in a document) in postings lists.
4. Boolean queries just retrieve a set of matching documents, but commonly we wish to have an effective method to order (or “rank”) the returned results. This requires having a mechanism for determining a document score which encapsulates how good a match a document is for a query.

TERM FREQUENCY

With these additional ideas, we will have seen most of the basic technology that supports ad hoc searching over unstructured information. Ad hoc searching over documents has recently conquered the world, powering not only web search engines but the kind of unstructured search that lies behind the large eCommerce websites. Although the main web search engines differ by emphasizing free text querying, most of the basic issues and technologies of indexing and querying remain the same, as we will see in later chapters. Moreover, over time, web search engines have added at least partial implementations of some of the most popular operators from extended Boolean models: phrase search is especially popular and most have a very partial implementation of Boolean operators. Nevertheless, while these options are liked by expert searchers, they are little used by most people and are not the main focus in work on trying to improve web search engine performance.

**Exercise 1.12**[\star]

Write a query using Westlaw syntax which would find any of the words professor, teacher, or lecturer in the same sentence as a form of the verb explain.

Exercise 1.13[\star]

Try using the Boolean search features on a couple of major web search engines. For instance, choose a word, such as burglar, and submit the queries (i) burglar, (ii) burglar AND burglar, and (iii) burglar OR burglar. Look at the estimated number of results and top hits. Do they make sense in terms of Boolean logic? Often they haven't for major search engines. Can you make sense of what is going on? What about if you try different words? For example, query for (i) knight, (ii) conquer, and then (iii) knight OR conquer. What bound should the number of results from the first two queries place on the third query? Is this bound observed?

1.5 References and further reading

The practical pursuit of computerized information retrieval began in the late 1940s ([Cleverdon 1991](#), [Liddy 2005](#)). A great increase in the production of scientific literature, much in the form of less formal technical reports rather than traditional journal articles, coupled with the availability of computers, led to interest in automatic document retrieval. However, in those days, document retrieval was always based on author, title, and keywords; full-text search came much later.

The article of [Bush \(1945\)](#) provided lasting inspiration for the new field:

"Consider a future device for individual use, which is a sort of mechanized private file and library. It needs a name, and, to coin one at random, 'memex' will do. A memex is a device in which an individual stores all his books, records, and communications, and which is mechanized so that it may be consulted with exceeding speed and flexibility. It is an enlarged intimate supplement to his memory."

The term *Information Retrieval* was coined by Calvin Mooers in 1948/1950 ([Mooers 1950](#)).

In 1958, much newspaper attention was paid to demonstrations at a conference (see [Taube and Wooster 1958](#)) of IBM "auto-indexing" machines, based primarily on the work of H. P. Luhn. Commercial interest quickly gravitated towards Boolean retrieval systems, but the early years saw a heady debate over various disparate technologies for retrieval systems. For example [Mooers \(1961\)](#) dissented:

"It is a common fallacy, underwritten at this date by the investment of several million dollars in a variety of retrieval hardware, that the algebra of George Boole (1847) is the appropriate formalism for retrieval system design. This view is as widely and uncritically accepted as it is wrong."

The observation of AND vs. OR giving you opposite extremes in a precision/recall tradeoff, but not the middle ground comes from ([Lee and Fox 1988](#)).

The book (Witten et al. 1999) is the standard reference for an in-depth comparison of the space and time efficiency of the inverted index versus other possible data structures; a more succinct and up-to-date presentation appears in Zobel and Moffat (2006). We further discuss several approaches in Chapter 5.

REGULAR EXPRESSIONS

Friedl (2006) covers the practical usage of *regular expressions* for searching. The underlying computer science appears in (Hopcroft et al. 2000).

2 *The term vocabulary and postings lists*

Recall the major steps in inverted index construction:

1. Collect the documents to be indexed.
2. Tokenize the text.
3. Do linguistic preprocessing of tokens.
4. Index the documents that each term occurs in.

In this chapter we first briefly mention how the basic unit of a document can be defined and how the character sequence that it comprises is determined (Section 2.1). We then examine in detail some of the substantive linguistic issues of tokenization and linguistic preprocessing, which determine the vocabulary of terms which a system uses (Section 2.2). Tokenization is the process of chopping character streams into tokens, while linguistic preprocessing then deals with building equivalence classes of tokens which are the set of terms that are indexed. Indexing itself is covered in Chapters 1 and 4. Then we return to the implementation of postings lists. In Section 2.3, we examine an extended postings list data structure that supports faster querying, while Section 2.4 covers building postings data structures suitable for handling phrase and proximity queries, of the sort that commonly appear in both extended Boolean models and on the web.

2.1 Document delineation and character sequence decoding

2.1.1 Obtaining the character sequence in a document

Digital documents that are the input to an indexing process are typically bytes in a file or on a web server. The first step of processing is to convert this byte sequence into a linear sequence of characters. For the case of plain English text in ASCII encoding, this is trivial. But often things get much more

complex. The sequence of characters may be encoded by one of various single byte or multibyte encoding schemes, such as Unicode UTF-8, or various national or vendor-specific standards. We need to determine the correct encoding. This can be regarded as a machine learning classification problem, as discussed in Chapter 13,¹ but is often handled by heuristic methods, user selection, or by using provided document metadata. Once the encoding is determined, we decode the byte sequence to a character sequence. We might save the choice of encoding because it gives some evidence about what language the document is written in.

The characters may have to be decoded out of some binary representation like Microsoft Word DOC files and/or a compressed format such as zip files. Again, we must determine the document format, and then an appropriate decoder has to be used. Even for plain text documents, additional decoding may need to be done. In XML documents (Section 10.1, page 197), character entities, such as &;, need to be decoded to give the correct character, namely & for &.. Finally, the textual part of the document may need to be extracted out of other material that will not be processed. This might be the desired handling for XML files, if the markup is going to be ignored; we would almost certainly want to do this with postscript or PDF files. We will not deal further with these issues in this book, and will assume henceforth that our documents are a list of characters. Commercial products usually need to support a broad range of document types and encodings, since users want things to just work with their data as is. Often, they just think of documents as text inside applications and are not even aware of how it is encoded on disk. This problem is usually solved by licensing a software library that handles decoding document formats and character encodings.

The idea that text is a linear sequence of characters is also called into question by some writing systems, such as Arabic, where text takes on some two dimensional and mixed order characteristics, as shown in Figures 2.1 and 2.2. But, despite some complicated writing system conventions, there is an underlying sequence of sounds being represented and hence an essentially linear structure remains, and this is what is represented in the digital representation of Arabic, as shown in Figure 2.1.

2.1.2 Choosing a document unit

DOCUMENT UNIT

The next phase is to determine what the *document unit* for indexing is. Thus far we have assumed that documents are fixed units for the purposes of indexing. For example, we take each file in a folder as a document. But there

1. A classifier is a function that takes objects of some sort and assigns them to one of a number of distinct classes (see Chapter 13). Usually classification is done by machine learning methods such as probabilistic models, but it can also be done by hand-written rules.

كِتَابٌ ← ا بَ ← كِتابٌ
 un b ā t i k
 /kitābun/ ‘a book’

► **Figure 2.1** An example of a vocalized Modern Standard Arabic word. The writing is from right to left and letters undergo complex mutations as they are combined. The representation of short vowels (here, /i/ and /u/) and the final /n/ (nunation) departs from strict linearity by being represented as diacritics above and below letters. Nevertheless, the represented text is still clearly a linear ordering of characters representing sounds. Full vocalization, as here, normally appears only in the Koran and children’s books. Day-to-day text is unvocalized (short vowels are not represented but the letter for ā would still appear) or partially vocalized, with short vowels inserted in places where the writer perceives ambiguities. These choices add further complexities to indexing.

استقللت الجزائر في سنة 1962 بعد 132 عاما من الاحتلال الفرنسي.

← → ← → ← START

‘Algeria achieved its independence in 1962 after 132 years of French occupation.’

► **Figure 2.2** The conceptual linear order of characters is not necessarily the order that you see on the page. In languages that are written right-to-left, such as Hebrew and Arabic, it is quite common to also have left-to-right text interspersed, such as numbers and dollar amounts. With modern Unicode representation concepts, the order of characters in files matches the conceptual order, and the reversal of displayed characters is handled by the rendering system, but this may not be true for documents in older encodings.

are many cases in which you might want to do something different. A traditional Unix (mbox-format) email file stores a sequence of email messages (an email folder) in one file, but you might wish to regard each email message as a separate document. Many email messages now contain attached documents, and you might then want to regard the email message and each contained attachment as separate documents. If an email message has an attached zip file, you might want to decode the zip file and regard each file it contains as a separate document. Going in the opposite direction, various pieces of web software (such as *latex2html*) take things that you might regard as a single document (e.g., a Powerpoint file or a L^AT_EX document) and split them into separate HTML pages for each slide or subsection, stored as separate files. In these cases, you might want to combine multiple files into a single document.

INDEXING
GRANULARITY

More generally, for very long documents, the issue of indexing *granularity* arises. For a collection of books, it would usually be a bad idea to index an

entire book as a document. A search for Chinese toys might bring up a book that mentions China in the first chapter and toys in the last chapter, but this does not make it relevant to the query. Instead, we may well wish to index each chapter or paragraph as a mini-document. Matches are then more likely to be relevant, and since the documents are smaller it will be much easier for the user to find the relevant passages in the document. But why stop there? We could treat individual sentences as mini-documents. It becomes clear that there is a precision/recall tradeoff here. If the units get too small, we are likely to miss important passages because terms were distributed over several mini-documents, while if units are too large we tend to get spurious matches and the relevant information is hard for the user to find.

The problems with large document units can be alleviated by use of explicit or implicit proximity search (Sections 2.4.2 and 7.2.2), and the trade-offs in resulting system performance that we are hinting at are discussed in Chapter 8. The issue of index granularity, and in particular a need to simultaneously index documents at multiple levels of granularity, appears prominently in XML retrieval, and is taken up again in Chapter 10. An IR system should be designed to offer choices of granularity. For this choice to be made well, the person who is deploying the system must have a good understanding of the document collection, the users, and their likely information needs and usage patterns. For now, we will henceforth assume that a suitable size document unit has been chosen, together with an appropriate way of dividing or aggregating files, if needed.

2.2 Determining the vocabulary of terms

2.2.1 Tokenization

Given a character sequence and a defined document unit, tokenization is the task of chopping it up into pieces, called *tokens*, perhaps at the same time throwing away certain characters, such as punctuation. Here is an example of tokenization:

Input: Friends, Romans, Countrymen, lend me your ears;							
Output:	Friends	Romans	Countrymen	lend	me	your	ears

These tokens are often loosely referred to as terms or words, but it is sometimes important to make a type/token distinction. A *token* is an instance of a sequence of characters in some particular document that are grouped together as a useful semantic unit for processing. A *type* is the class of all tokens containing the same character sequence. A *term* is a (perhaps normalized) type that is included in the IR system's dictionary. The set of index terms could be entirely distinct from the tokens, for instance, they could be

semantic identifiers in a taxonomy, but in practice in modern IR systems they are strongly related to the tokens in the document. However, rather than being exactly the tokens that appear in the document, they are usually derived from them by various normalization processes which are discussed in Section 2.2.3.² For example, if the document to be indexed is *to sleep perchance to dream*, then there are 5 tokens, but only 4 types (since there are 2 instances of *to*). However, if *to* is omitted from the index (as a stop word, see Section 2.2.2 (page 27)), then there will be only 3 terms: *sleep*, *perchance*, and *dream*.

The major question of the tokenization phase is what are the correct tokens to use? In this example, it looks fairly trivial: you chop on whitespace and throw away punctuation characters. This is a starting point, but even for English there are a number of tricky cases. For example, what do you do about the various uses of the apostrophe for possession and contractions?

Mr. O'Neill thinks that the boys' stories about Chile's capital aren't amusing.

For O'Neill, which of the following is the desired tokenization?

neill
oneill
o'neill
o' neill
o neill?

And for *aren't*, is it:

aren't
arent
are n't
aren t?

A simple strategy is to just split on all non-alphanumeric characters, but while

o	neill
---	-------

 looks okay,

aren	t
------	---

 looks intuitively bad. For all of them, the choices determine which Boolean queries will match. A query of *neill* AND *capital* will match in three cases but not the other two. In how many cases would a query of *o'neill* AND *capital* match? If no preprocessing of a query is done, then it would match in only one of the five cases. For either

2. That is, as defined here, tokens that are not indexed (stop words) are not terms, and if multiple tokens are collapsed together via normalization, they are indexed as one term, under the normalized form. However, we later relax this definition when discussing classification and clustering in Chapters 13–18, where there is no index. In these chapters, we drop the requirement of inclusion in the dictionary. A *term* means a normalized word.

LANGUAGE
IDENTIFICATION

Boolean or free text queries, you always want to do the exact same tokenization of document and query words, generally by processing queries with the same tokenizer. This guarantees that a sequence of characters in a text will always match the same sequence typed in a query.³

These issues of tokenization are language-specific. It thus requires the language of the document to be known. *Language identification* based on classifiers that use short character subsequences as features is highly effective; most languages have distinctive signature patterns (see page 46 for references).

For most languages and particular domains within them there are unusual specific tokens that we wish to recognize as terms, such as the programming languages C++ and C#, aircraft names like B-52, or a T.V. show name such as M*A*S*H – which is sufficiently integrated into popular culture that you find usages such as *M*A*S*H-style hospitals*. Computer technology has introduced new types of character sequences that a tokenizer should probably tokenize as a single token, including email addresses (`jblack@mail.yahoo.com`), web URLs (<http://stuff.big.com/new/specials.html>), numeric IP addresses (142.32.48.231), package tracking numbers (1Z9999W99845399981), and more. One possible solution is to omit from indexing tokens such as monetary amounts, numbers, and URLs, since their presence greatly expands the size of the vocabulary. However, this comes at a large cost in restricting what people can search for. For instance, people might want to search in a bug database for the line number where an error occurs. Items such as the date of an email, which have a clear semantic type, are often indexed separately as document metadata (see Section 6.1, page 110).

HYPHENS

In English, *hyphenation* is used for various purposes ranging from splitting up vowels in words (*co-education*) to joining nouns as names (*Hewlett-Packard*) to a copyediting device to show word grouping (*the hold-him-back-and-drag-him-away maneuver*). It is easy to feel that the first example should be regarded as one token (and is indeed more commonly written as just *coeducation*), the last should be separated into words, and that the middle case is unclear. Handling hyphens automatically can thus be complex: it can either be done as a classification problem, or more commonly by some heuristic rules, such as allowing short hyphenated prefixes on words, but not longer hyphenated forms.

Conceptually, splitting on white space can also split what should be regarded as a single token. This occurs most commonly with names (*San Francisco, Los Angeles*) but also with borrowed foreign phrases (*au fait*) and com-

3. For the free text case, this is straightforward. The Boolean case is more complex: this tokenization may produce multiple terms from one query word. This can be handled by combining the terms with an AND or as a phrase query (see Section 2.4, page 39). It is harder for a system to handle the opposite case where the user entered as two terms something that was tokenized together in the document processing.

pounds that are sometimes written as a single word and sometimes space separated (such as *white space* vs. *whitespace*). Other cases with internal spaces that we might wish to regard as a single token include phone numbers ((800) 234-2333) and dates (Mar 11, 1983). Splitting tokens on spaces can cause bad retrieval results, for example, if a search for York University mainly returns documents containing *New York University*. The problems of hyphens and non-separating whitespace can even interact. Advertisements for air fares frequently contain items like *San Francisco-Los Angeles*, where simply doing whitespace splitting would give unfortunate results. In such cases, issues of tokenization interact with handling phrase queries (which we discuss in Section 2.4 (page 39)), particularly if we would like queries for all of *lowercase*, *lower-case* and *lower case* to return the same results. The last two can be handled by splitting on hyphens and using a phrase index. Getting the first case right would depend on knowing that it is sometimes written as two words and also indexing it in this way. One effective strategy in practice, which is used by some Boolean retrieval systems such as Westlaw and Lexis-Nexis (Example 1.1), is to encourage users to enter hyphens wherever they may be possible, and whenever there is a hyphenated form, the system will generalize the query to cover all three of the one word, hyphenated, and two word forms, so that a query for *over-eager* will search for *over-eager* OR “*over eager*” OR *overeager*. However, this strategy depends on user training, since if you query using either of the other two forms, you get no generalization.

Each new language presents some new issues. For instance, French has a variant use of the apostrophe for a reduced definite article ‘the’ before a word beginning with a vowel (e.g., *l’ensemble*) and has some uses of the hyphen with postposed clitic pronouns in imperatives and questions (e.g., *donne-moi* ‘give me’). Getting the first case correct will affect the correct indexing of a fair percentage of nouns and adjectives: you would want documents mentioning both *l’ensemble* and *un ensemble* to be indexed under *ensemble*. Other languages make the problem harder in new ways. German writes *compound nouns* without spaces (e.g., *Computerlinguistik* ‘computational linguistics’; *Lebensversicherungsgesellschaftsangestellter* ‘life insurance company employee’). Retrieval systems for German greatly benefit from the use of a *compound-splitter* module, which is usually implemented by seeing if a word can be subdivided into multiple words that appear in a vocabulary. This phenomenon reaches its limit case with major East Asian Languages (e.g., Chinese, Japanese, Korean, and Thai), where text is written without any spaces between words. An example is shown in Figure 2.3. One approach here is to perform *word segmentation* as prior linguistic processing. Methods of word segmentation vary from having a large vocabulary and taking the longest vocabulary match with some heuristics for unknown words to the use of machine learning sequence models, such as hidden Markov models or conditional random fields, trained over hand-segmented words (see the references

COMPOUNDS

COMPOUND-SPLITTER

WORD SEGMENTATION

莎拉波娃现在居住在美国东南部的佛罗里达。今年4月9日，莎拉波娃在美国第一大城市纽约度过了18岁生日。生日派对上，莎拉波娃露出了甜美的微笑。

► **Figure 2.3** The standard unsegmented form of Chinese text using the simplified characters of mainland China. There is no whitespace between words, not even between sentences – the apparent space after the Chinese period (。) is just a typographical illusion caused by placing the character on the left side of its square box. The first sentence is just words in Chinese characters with no spaces between them. The second and third sentences include Arabic numerals and punctuation breaking up the Chinese characters.

和尚

► **Figure 2.4** Ambiguities in Chinese word segmentation. The two characters can be treated as one word meaning ‘monk’ or as a sequence of two words meaning ‘and’ and ‘still’.

a	an	and	are	as	at	be	by	for	from
has	he	in	is	it	its	of	on	that	the
to	was	were	will	with					

► **Figure 2.5** A stop list of 25 semantically non-selective words which are common in Reuters-RCV1.

in Section 2.5). Since there are multiple possible segmentations of character sequences (see Figure 2.4), all such methods make mistakes sometimes, and so you are never guaranteed a consistent unique tokenization. The other approach is to abandon word-based indexing and to do all indexing via just short subsequences of characters (character k -grams), regardless of whether particular sequences cross word boundaries or not. Three reasons why this approach is appealing are that an individual Chinese character is more like a syllable than a letter and usually has some semantic content, that most words are short (the commonest length is 2 characters), and that, given the lack of standardization of word breaking in the writing system, it is not always clear where word boundaries should be placed anyway. Even in English, some cases of where to put word boundaries are just orthographic conventions – think of *notwithstanding* vs. *not to mention* or *into* vs. *on to* – but people are educated to write the words with consistent use of spaces.

2.2.2 Dropping common terms: stop words

STOP WORDS
COLLECTION
FREQUENCY

STOP LIST

Sometimes, some extremely common words which would appear to be of little value in helping select documents matching a user need are excluded from the vocabulary entirely. These words are called *stop words*. The general strategy for determining a stop list is to sort the terms by *collection frequency* (the total number of times each term appears in the document collection), and then to take the most frequent terms, often hand-filtered for their semantic content relative to the domain of the documents being indexed, as a *stop list*, the members of which are then discarded during indexing. An example of a stop list is shown in Figure 2.5. Using a stop list significantly reduces the number of postings that a system has to store; we will present some statistics on this in Chapter 5 (see Table 5.1, page 87). And a lot of the time not indexing stop words does little harm: keyword searches with terms like the and by don't seem very useful. However, this is not true for phrase searches. The phrase query "President of the United States", which contains two stop words, is more precise than President AND "United States". The meaning of flights to London is likely to be lost if the word to is stopped out. A search for Vannevar Bush's article *As we may think* will be difficult if the first three words are stopped out, and the system searches simply for documents containing the word think. Some special query types are disproportionately affected. Some song titles and well known pieces of verse consist entirely of words that are commonly on stop lists (*To be or not to be, Let It Be, I don't want to be, ...*).

The general trend in IR systems over time has been from standard use of quite large stop lists (200–300 terms) to very small stop lists (7–12 terms) to no stop list whatsoever. Web search engines generally do not use stop lists. Some of the design of modern IR systems has focused precisely on how we can exploit the statistics of language so as to be able to cope with common words in better ways. We will show in Section 5.3 (page 95) how good compression techniques greatly reduce the cost of storing the postings for common words. Section 6.2.1 (page 117) then discusses how standard term weighting leads to very common words having little impact on document rankings. Finally, Section 7.1.5 (page 140) shows how an IR system with impact-sorted indexes can terminate scanning a postings list early when weights get small, and hence common words do not cause a large additional processing cost for the average query, even though postings lists for stop words are very long. So for most modern IR systems, the additional cost of including stop words is not that big – neither in terms of index size nor in terms of query processing time.

Query term	Terms in documents that should be matched
Windows	Windows
windows	Windows, windows, window
window	window, windows

► **Figure 2.6** An example of how asymmetric expansion of query terms can usefully model users' expectations.

2.2.3 Normalization (equivalence classing of terms)

Having broken up our documents (and also our query) into tokens, the easy case is if tokens in the query just match tokens in the token list of the document. However, there are many cases when two character sequences are not quite the same but you would like a match to occur. For instance, if you search for *USA*, you might hope to also match documents containing *U.S.A.*

TOKEN
NORMALIZATION
EQUIVALENCE CLASSES

Token normalization is the process of canonicalizing tokens so that matches occur despite superficial differences in the character sequences of the tokens.⁴ The most standard way to normalize is to implicitly create *equivalence classes*, which are normally named after one member of the set. For instance, if the tokens *anti-discriminatory* and *antidiscriminatory* are both mapped onto the term *antidiscriminatory*, in both the document text and queries, then searches for one term will retrieve documents that contain either.

The advantage of just using mapping rules that remove characters like hyphens is that the equivalence classing to be done is implicit, rather than being fully calculated in advance: the terms that happen to become identical as the result of these rules are the equivalence classes. It is only easy to write rules of this sort that remove characters. Since the equivalence classes are implicit, it is not obvious when you might want to add characters. For instance, it would be hard to know to turn *antidiscriminatory* into *anti-discriminatory*.

An alternative to creating equivalence classes is to maintain relations between unnormalized tokens. This method can be extended to hand-constructed lists of synonyms such as *car* and *automobile*, a topic we discuss further in Chapter 9. These term relationships can be achieved in two ways. The usual way is to index unnormalized tokens and to maintain a query expansion list of multiple vocabulary entries to consider for a certain query term. A query term is then effectively a disjunction of several postings lists. The alternative is to perform the expansion during index construction. When the document contains *automobile*, we index it under *car* as well (and, usually, also vice-versa). Use of either of these methods is considerably less efficient than equivalence classing, as there are more postings to store and merge. The first

4. It is also often referred to as *term normalization*, but we prefer to reserve the name *term* for the output of the normalization process.

method adds a query expansion dictionary and requires more processing at query time, while the second method requires more space for storing postings. Traditionally, expanding the space required for the postings lists was seen as more disadvantageous, but with modern storage costs, the increased flexibility that comes from distinct postings lists is appealing.

These approaches are more flexible than equivalence classes because the expansion lists can overlap while not being identical. This means there can be an asymmetry in expansion. An example of how such an asymmetry can be exploited is shown in Figure 2.6: if the user enters *windows*, we wish to allow matches with the capitalized *Windows* operating system, but this is not plausible if the user enters *window*, even though it is plausible for this query to also match lowercase *windows*.

The best amount of equivalence classing or query expansion to do is a fairly open question. Doing some definitely seems a good idea. But doing a lot can easily have unexpected consequences of broadening queries in unintended ways. For instance, equivalence-classing *U.S.A.* and *USA* to the latter by deleting periods from tokens might at first seem very reasonable, given the prevalent pattern of optional use of periods in acronyms. However, if I put in as my query term *C.A.T.*, I might be rather upset if it matches every appearance of the word *cat* in documents.⁵

Below we present some of the forms of normalization that are commonly employed and how they are implemented. In many cases they seem helpful, but they can also do harm. In fact, you can worry about many details of equivalence classing, but it often turns out that providing processing is done consistently to the query and to documents, the fine details may not have much aggregate effect on performance.

Accents and diacritics. Diacritics on characters in English have a fairly marginal status, and we might well want *cliché* and *cliche* to match, or *naïve* and *naive*. This can be done by normalizing tokens to remove diacritics. In many other languages, diacritics are a regular part of the writing system and distinguish different sounds. Occasionally words are distinguished only by their accents. For instance, in Spanish, *peña* is ‘a cliff’, while *pena* is ‘sorrow’. Nevertheless, the important question is usually not prescriptive or linguistic but is a question of how users are likely to write queries for these words. In many cases, users will enter queries for words without diacritics, whether for reasons of speed, laziness, limited software, or habits born of the days when it was hard to use non-ASCII text on many computer systems. In these cases, it might be best to equate all words to a form without diacritics.

5. At the time we wrote this chapter (Aug. 2005), this was actually the case on Google: the top result for the query *C.A.T.* was a site about cats, the Cat Fanciers Web Site <http://www.fanciers.com/>.

CASE-FOLDING

Capitalization/case-folding. A common strategy is to do *case-folding* by reducing all letters to lower case. Often this is a good idea: it will allow instances of *Automobile* at the beginning of a sentence to match with a query of *automobile*. It will also help on a web search engine when most of your users type in *ferrari* when they are interested in a *Ferrari* car. On the other hand, such case folding can equate words that might better be kept apart. Many proper nouns are derived from common nouns and so are distinguished only by case, including companies (*General Motors*, *The Associated Press*), government organizations (*the Fed* vs. *fed*) and person names (*Bush*, *Black*). We already mentioned an example of unintended query expansion with acronyms, which involved not only acronym normalization (*C.A.T.* → *CAT*) but also case-folding (*CAT* → *cat*).

TRUECASING

For English, an alternative to making every token lowercase is to just make some tokens lowercase. The simplest heuristic is to convert to lowercase words at the beginning of a sentence and all words occurring in a title that is all uppercase or in which most or all words are capitalized. These words are usually ordinary words that have been capitalized. Mid-sentence capitalized words are left as capitalized (which is usually correct). This will mostly avoid case-folding in cases where distinctions should be kept apart. The same task can be done more accurately by a machine learning sequence model which uses more features to make the decision of when to case-fold. This is known as *truecasing*. However, trying to get capitalization right in this way probably doesn't help if your users usually use lowercase regardless of the correct case of words. Thus, lowercasing everything often remains the most practical solution.

Other issues in English. Other possible normalizations are quite idiosyncratic and particular to English. For instance, you might wish to equate *ne'er* and *never* or the British spelling *colour* and the American spelling *color*. Dates, times and similar items come in multiple formats, presenting additional challenges. You might wish to collapse together *3/12/91* and *Mar. 12, 1991*. However, correct processing here is complicated by the fact that in the U.S., *3/12/91* is *Mar. 12, 1991*, whereas in Europe it is *3 Dec 1991*.

Other languages. English has maintained a dominant position on the WWW; approximately 60% of web pages are in English (Gerrand 2007). But that still leaves 40% of the web, and the non-English portion might be expected to grow over time, since less than one third of Internet users and less than 10% of the world's population primarily speak English. And there are signs of change: Sifry (2007) reports that only about one third of blog posts are in English.

Other languages again present distinctive issues in equivalence classing.

ノーベル平和賞を受賞したワンガリ・マータイさんが名誉会長を務めるMOTTAINIキャンペーンの一環として、毎日新聞社とマガジンハウスは「私の、もったいない」を募集します。皆様が日ごろ「もったいない」と感じて実践していることや、それにまつわるエピソードを800字以内の文章にまとめ、簡単な写真、イラスト、図などを添えて10月20日までにお送りください。大賞受賞者には、50万円相当の旅行券とエコ製品2点の副賞が贈られます。

► **Figure 2.7** Japanese makes use of multiple intermingled writing systems and, like Chinese, does not segment words. The text is mainly Chinese characters with the hiragana syllabary for inflectional endings and function words. The part in latin letters is actually a Japanese expression, but has been taken up as the name of an environmental campaign by 2004 Nobel Peace Prize winner Wangari Maathai. His name is written using the katakana syllabary in the middle of the first line. The first four characters of the final line express a monetary amount that we would want to match with ¥500,000 (500,000 Japanese yen).

The French word for *the* has distinctive forms based not only on the gender (masculine or feminine) and number of the following noun, but also depending on whether the following word begins with a vowel: *le, la, l', les*. We may well wish to equivalence class these various forms of *the*. German has a convention whereby vowels with an umlaut can be rendered instead as a two vowel digraph. We would want to treat *Schütze* and *Schuetze* as equivalent.

Japanese is a well-known difficult writing system, as illustrated in Figure 2.7. Modern Japanese is standardly an intermingling of multiple alphabets, principally Chinese characters, two syllabaries (hiragana and katakana) and western characters (Latin letters, Arabic numerals, and various symbols). While there are strong conventions and standardization through the education system over the choice of writing system, in many cases the same word can be written with multiple writing systems. For example, a word may be written in katakana for emphasis (somewhat like italics). Or a word may sometimes be written in hiragana and sometimes in Chinese characters. Successful retrieval thus requires complex equivalence classing across the writing systems. In particular, an end user might commonly present a query entirely in hiragana, because it is easier to type, just as Western end users commonly use all lowercase.

Document collections being indexed can include documents from many different languages. Or a single document can easily contain text from multiple languages. For instance, a French email might quote clauses from a contract document written in English. Most commonly, the language is detected and language-particular tokenization and normalization rules are applied at a predetermined granularity, such as whole documents or individual paragraphs, but this still will not correctly deal with cases where language changes occur for brief quotations. When document collections contain mul-

tiple languages, a single index may have to contain terms of several languages. One option is to run a language identification classifier on documents and then to tag terms in the vocabulary for their language. Or this tagging can simply be omitted, since it is relatively rare for the exact same character sequence to be a word in different languages.

When dealing with foreign or complex words, particularly foreign names, the spelling may be unclear or there may be variant transliteration standards giving different spellings (for example, *Chebyshev* and *Tchebycheff* or *Beijing* and *Peking*). One way of dealing with this is to use heuristics to equivalence class or expand terms with phonetic equivalents. The traditional and best known such algorithm is the Soundex algorithm, which we cover in Section 3.4 (page 63).

2.2.4 Stemming and lemmatization

For grammatical reasons, documents are going to use different forms of a word, such as *organize*, *organizes*, and *organizing*. Additionally, there are families of derivationally related words with similar meanings, such as *democracy*, *democratic*, and *democratization*. In many situations, it seems as if it would be useful for a search for one of these words to return documents that contain another word in the set.

The goal of both stemming and lemmatization is to reduce inflectional forms and sometimes derivationally related forms of a word to a common base form. For instance:

am, are, is ⇒ be
car, cars, car's, cars' ⇒ car

The result of this mapping of text will be something like:

the boy's cars are different colors ⇒
the boy car be differ color

STEMMING	However, the two words differ in their flavor. <i>Stemming</i> usually refers to a crude heuristic process that chops off the ends of words in the hope of achieving this goal correctly most of the time, and often includes the removal of derivational affixes. <i>Lemmatization</i> usually refers to doing things properly with the use of a vocabulary and morphological analysis of words, normally aiming to remove inflectional endings only and to return the base or dictionary form of a word, which is known as the <i>lemma</i> . If confronted with the token <i>saw</i> , stemming might return just <i>s</i> , whereas lemmatization would attempt to return either <i>see</i> or <i>saw</i> depending on whether the use of the token was as a verb or a noun. The two may also differ in that stemming most commonly collapses derivationally related words, whereas lemmatization commonly only collapses the different inflectional forms of a lemma.
LEMMATIZATION	
LEMMA	

Linguistic processing for stemming or lemmatization is often done by an additional plug-in component to the indexing process, and a number of such components exist, both commercial and open-source.

PORTER STEMMER

The most common algorithm for stemming English, and one that has repeatedly been shown to be empirically very effective, is *Porter's algorithm* (Porter 1980). The entire algorithm is too long and intricate to present here, but we will indicate its general nature. Porter's algorithm consists of 5 phases of word reductions, applied sequentially. Within each phase there are various conventions to select rules, such as selecting the rule from each rule group that applies to the longest suffix. In the first phase, this convention is used with the following rule group:

(2.1)	Rule	Example
SSES	→ SS	caresses → caress
IES	→ I	ponies → poni
SS	→ SS	caress → caress
S	→	cats → cat

Many of the later rules use a concept of the *measure* of a word, which loosely checks the number of syllables to see whether a word is long enough that it is reasonable to regard the matching portion of a rule as a suffix rather than as part of the stem of a word. For example, the rule:

($m > 1$) EMENT →

would map *replacement* to *replac*, but not *cement* to *c*. The official site for the Porter Stemmer is:

<http://www.tartarus.org/~martin/PorterStemmer/>

Other stemmers exist, including the older, one-pass Lovins stemmer (Lovins 1968), and newer entrants like the Paice/Husk stemmer (Paice 1990); see:

<http://www.cs.waikato.ac.nz/~eibe/stemmers/>

<http://www.comp.lancs.ac.uk/computing/research/stemming/>

Figure 2.8 presents an informal comparison of the different behaviors of these stemmers. Stemmers use language-specific rules, but they require less knowledge than a lemmatizer, which needs a complete vocabulary and morphological analysis to correctly lemmatize words. Particular domains may also require special stemming rules. However, the exact stemmed form does not matter, only the equivalence classes it forms.

LEMMATIZER

Rather than using a stemmer, you can use a *lemmatizer*, a tool from Natural Language Processing which does full morphological analysis to accurately identify the lemma for each word. Doing full morphological analysis produces at most very modest benefits for retrieval. It is hard to say more,

Sample text: Such an analysis can reveal features that are not easily visible from the variations in the individual genes and can lead to a picture of expression that is more biologically transparent and accessible to interpretation

Lovins stemmer: such an analys can reve featur that ar not eas vis from th vari in th individu gen and can lead to a pictur of expres that is mor biolog transpar and acces to interpres

Porter stemmer: such an analysi can reveal featur that ar not easili visibl from the variat in the individu gene and can lead to a pictur of express that is more biolog transpar and access to interpret

Paice stemmer: such an analys can rev feat that are not easy vis from the vary in the individ gen and can lead to a pict of express that is mor biolog transp and access to interpret

► **Figure 2.8** A comparison of three stemming algorithms on a sample text.

because either form of normalization tends not to improve English information retrieval performance in aggregate – at least not by very much. While it helps a lot for some queries, it equally hurts performance a lot for others. Stemming increases recall while harming precision. As an example of what can go wrong, note that the Porter stemmer stems all of the following words:

operate operating operates operation operative operatives operational

to oper. However, since *operate* in its various forms is a common verb, we would expect to lose considerable precision on queries such as the following with Porter stemming:

operational AND research
operating AND system
operative AND dentistry

For a case like this, moving to using a lemmatizer would not completely fix the problem because particular inflectional forms are used in particular collocations: a sentence with the words *operate* and *system* is not a good match for the query *operating AND system*. Getting better value from term normalization depends more on pragmatic issues of word use than on formal issues of linguistic morphology.

The situation is different for languages with much more morphology (such as Spanish, German, and Finnish). Results in the European CLEF evaluations have repeatedly shown quite large gains from the use of stemmers (and compound splitting for languages like German); see the references in Section 2.5.

**Exercise 2.1**

[*]

Are the following statements true or false?

- In a Boolean retrieval system, stemming never lowers precision.
- In a Boolean retrieval system, stemming never lowers recall.
- Stemming increases the size of the vocabulary.
- Stemming should be invoked at indexing time but not while processing a query.

Exercise 2.2

[*]

Suggest what normalized form should be used for these words (including the word itself as a possibility):

- 'Cos
- Shi'ite
- cont'd
- Hawai'i
- O'Rourke

Exercise 2.3

[*]

The following pairs of words are stemmed to the same form by the Porter stemmer. Which pairs would you argue shouldn't be conflated. Give your reasoning.

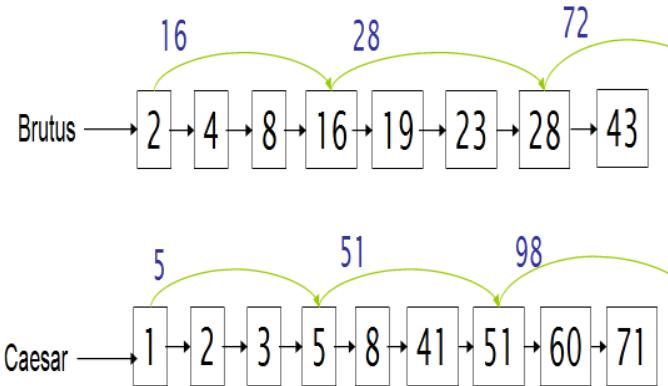
- abandon/abandonment
- absorbency/absorbent
- marketing/markets
- university/universe
- volume/volumes

Exercise 2.4

[*]

For the Porter stemmer rule group shown in (2.1):

- What is the purpose of including an identity rule such as SS → SS?
- Applying just this rule group, what will the following words be stemmed to?
circus canaries boss
- What rule should be added to correctly stem *pony*?
- The stemming for *ponies* and *pony* might seem strange. Does it have a deleterious effect on retrieval? Why or why not?



► **Figure 2.9** Postings lists with skip pointers. The postings intersection can use a skip pointer when the end point is still less than the item on the other list.

2.3 Faster postings list intersection via skip pointers

In the remainder of this chapter, we will discuss extensions to postings list data structures and ways to increase the efficiency of using postings lists. Recall the basic postings list intersection operation from Section 1.3 (page 10): we walk through the two postings lists simultaneously, in time linear in the total number of postings entries. If the list lengths are m and n , the intersection takes $O(m + n)$ operations. Can we do better than this? That is, empirically, can we usually process postings list intersection in sublinear time? We can, if the index isn't changing too fast.

SKIP LIST

One way to do this is to use a *skip list* by augmenting postings lists with skip pointers (at indexing time), as shown in Figure 2.9. Skip pointers are effectively shortcuts that allow us to avoid processing parts of the postings list that will not figure in the search results. The two questions are then where to place skip pointers and how to do efficient merging using skip pointers.

Consider first efficient merging, with Figure 2.9 as an example. Suppose we've stepped through the lists in the figure until we have matched [8] on each list and moved it to the results list. We advance both pointers, giving us [16] on the upper list and [41] on the lower list. The smallest item is then the element [16] on the top list. Rather than simply advancing the upper pointer, we first check the skip list pointer and note that 28 is also less than 41. Hence we can follow the skip list pointer, and then we advance the upper pointer to [28]. We thus avoid stepping to [19] and [23] on the upper list. A number of variant versions of postings list intersection with skip pointers is possible depending on when exactly you check the skip pointer. One version is shown

```

INTERSECTWITHSKIPS( $p_1, p_2$ )
1  $answer \leftarrow \langle \rangle$ 
2 while  $p_1 \neq \text{NIL}$  and  $p_2 \neq \text{NIL}$ 
3 do if  $docID(p_1) = docID(p_2)$ 
4   then ADD( $answer, docID(p_1)$ )
5      $p_1 \leftarrow next(p_1)$ 
6      $p_2 \leftarrow next(p_2)$ 
7   else if  $docID(p_1) < docID(p_2)$ 
8     then if  $hasSkip(p_1)$  and  $(docID(skip(p_1)) \leq docID(p_2))$ 
9       then while  $hasSkip(p_1)$  and  $(docID(skip(p_1)) \leq docID(p_2))$ 
10      do  $p_1 \leftarrow skip(p_1)$ 
11    else  $p_1 \leftarrow next(p_1)$ 
12    else if  $hasSkip(p_2)$  and  $(docID(skip(p_2)) \leq docID(p_1))$ 
13      then while  $hasSkip(p_2)$  and  $(docID(skip(p_2)) \leq docID(p_1))$ 
14        do  $p_2 \leftarrow skip(p_2)$ 
15    else  $p_2 \leftarrow next(p_2)$ 
16 return  $answer$ 

```

► **Figure 2.10** Postings lists intersection with skip pointers.

in Figure 2.10. Skip pointers will only be available for the original postings lists. For an intermediate result in a complex query, the call $hasSkip(p)$ will always return false. Finally, note that the presence of skip pointers only helps for AND queries, not for OR queries.

Where do we place skips? There is a tradeoff. More skips means shorter skip spans, and that we are more likely to skip. But it also means lots of comparisons to skip pointers, and lots of space storing skip pointers. Fewer skips means few pointer comparisons, but then long skip spans which means that there will be fewer opportunities to skip. A simple heuristic for placing skips, which has been found to work well in practice, is that for a postings list of length P , use \sqrt{P} evenly-spaced skip pointers. This heuristic can be improved upon; it ignores any details of the distribution of query terms.

Building effective skip pointers is easy if an index is relatively static; it is harder if a postings list keeps changing because of updates. A malicious deletion strategy can render skip lists ineffective.

Choosing the optimal encoding for an inverted index is an ever-changing game for the system builder, because it is strongly dependent on underlying computer technologies and their relative speeds and sizes. Traditionally, CPUs were slow, and so highly compressed techniques were not optimal. Now CPUs are fast and disk is slow, so reducing disk postings list size dominates. However, if you’re running a search engine with everything in mem-

ory then the equation changes again. We discuss the impact of hardware parameters on index construction time in Section 4.1 (page 68) and the impact of index size on system speed in Chapter 5.


Exercise 2.5

[★]

Why are skip pointers not useful for queries of the form $x \text{ OR } y$?

Exercise 2.6

[★]

We have a two-word query. For one term the postings list consists of the following 16 entries:

[4,6,10,12,14,16,18,20,22,32,47,81,120,122,157,180]

and for the other it is the one entry postings list:

[47].

Work out how many comparisons would be done to intersect the two postings lists with the following two strategies. Briefly justify your answers:

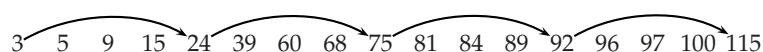
a. Using standard postings lists

b. Using postings lists stored with skip pointers, with a skip length of \sqrt{P} , as suggested in Section 2.3.

Exercise 2.7

[★]

Consider a postings intersection between this postings list, with skip pointers:



and the following intermediate result postings list (which hence has no skip pointers):

3 5 89 95 97 99 100 101

Trace through the postings intersection algorithm in Figure 2.10 (page 37).

- How often is a skip pointer followed (i.e., p_1 is advanced to $\text{skip}(p_1)$)?
- How many postings comparisons will be made by this algorithm while intersecting the two lists?
- How many postings comparisons would be made if the postings lists are intersected without the use of skip pointers?

2.4 Positional postings and phrase queries

PHRASE QUERIES

Many complex or technical concepts and many organization and product names are multiword compounds or phrases. We would like to be able to pose a query such as Stanford University by treating it as a phrase so that a sentence in a document like *The inventor Stanford Ovshinsky never went to university.* is not a match. Most recent search engines support a double quotes syntax ("stanford university") for *phrase queries*, which has proven to be very easily understood and successfully used by users. As many as 10% of web queries are phrase queries, and many more are implicit phrase queries (such as person names), entered without use of double quotes. To be able to support such queries, it is no longer sufficient for postings lists to be simply lists of documents that contain individual terms. In this section we consider two approaches to supporting phrase queries and their combination. A search engine should not only support phrase queries, but implement them efficiently. A related but distinct concept is term proximity weighting, where a document is preferred to the extent that the query terms appear close to each other in the text. This technique is covered in Section 7.2.2 (page 144) in the context of ranked retrieval.

BIWORD INDEX

2.4.1 Biword indexes

One approach to handling phrases is to consider every pair of consecutive terms in a document as a phrase. For example, the text *Friends, Romans, Countrymen* would generate the *biwords*:

```
friends romans
romans countrymen
```

In this model, we treat each of these biwords as a vocabulary term. Being able to process two-word phrase queries is immediate. Longer phrases can be processed by breaking them down. The query *stanford university palo alto* can be broken into the Boolean query on biwords:

```
"stanford university" AND "university palo" AND "palo alto"
```

This query could be expected to work fairly well in practice, but there can and will be occasional false positives. Without examining the documents, we cannot verify that the documents matching the above Boolean query do actually contain the original 4 word phrase.

Among possible queries, nouns and noun phrases have a special status in describing the concepts people are interested in searching for. But related nouns can often be divided from each other by various function words, in phrases such as *the abolition of slavery* or *renegotiation of the constitution*. These needs can be incorporated into the biword indexing model in the following

way. First, we tokenize the text and perform part-of-speech-tagging.⁶ We can then group terms into nouns, including proper nouns, (N) and function words, including articles and prepositions, (X), among other classes. Now deem any string of terms of the form NX*N to be an extended biword. Each such extended biword is made a term in the vocabulary. For example:

renegotiation	of	the	constitution
N	X	X	N

To process a query using such an extended biword index, we need to also parse it into N's and X's, and then segment the query into extended biwords, which can be looked up in the index.

This algorithm does not always work in an intuitively optimal manner when parsing longer queries into Boolean queries. Using the above algorithm, the query

cost overruns on a power plant

is parsed into

“cost overruns” AND “overruns power” AND “power plant”

whereas it might seem a better query to omit the middle biword. Better results can be obtained by using more precise part-of-speech patterns that define which extended biwords should be indexed.

PHRASE INDEX

The concept of a biword index can be extended to longer sequences of words, and if the index includes variable length word sequences, it is generally referred to as a *phrase index*. Indeed, searches for a single term are not naturally handled in a biword index (you would need to scan the dictionary for all biwords containing the term), and so we also need to have an index of single-word terms. While there is always a chance of false positive matches, the chance of a false positive match on indexed phrases of length 3 or more becomes very small indeed. But on the other hand, storing longer phrases has the potential to greatly expand the vocabulary size. Maintaining exhaustive phrase indexes for phrases of length greater than two is a daunting prospect, and even use of an exhaustive biword dictionary greatly expands the size of the vocabulary. However, towards the end of this section we discuss the utility of the strategy of using a partial phrase index in a compound indexing scheme.

6. Part of speech taggers classify words as nouns, verbs, etc. – or, in practice, often as finer-grained classes like “plural proper noun”. Many fairly accurate (c. 96% per-tag accuracy) part-of-speech taggers now exist, usually trained by machine learning methods on hand-tagged text. See, for instance, Manning and Schütze (1999, ch. 10).

```

to, 993427:
⟨ 1, 6: ⟨7, 18, 33, 72, 86, 231⟩;
  2, 5: ⟨1, 17, 74, 222, 255⟩;
  4, 5: ⟨8, 16, 190, 429, 433⟩;
  5, 2: ⟨363, 367⟩;
  7, 3: ⟨13, 23, 191⟩; ... ⟩

be, 178239:
⟨ 1, 2: ⟨17, 25⟩;
  4, 5: ⟨17, 191, 291, 430, 434⟩;
  5, 3: ⟨14, 19, 101⟩; ... ⟩

```

► **Figure 2.11** Positional index example. The word *to* has a document frequency 993,477, and occurs 6 times in document 1 at positions 7, 18, 33, etc.

2.4.2 Positional indexes

POSITIONAL INDEX

For the reasons given, a biword index is not the standard solution. Rather, a *positional index* is most commonly employed. Here, for each term in the vocabulary, we store postings of the form docID: ⟨position1, position2, ...⟩, as shown in Figure 2.11, where each position is a token index in the document. Each posting will also usually record the term frequency, for reasons discussed in Chapter 6.

To process a phrase query, you still need to access the inverted index entries for each distinct term. As before, you would start with the least frequent term and then work to further restrict the list of possible candidates. In the merge operation, the same general technique is used as before, but rather than simply checking that both terms are in a document, you also need to check that their positions of appearance in the document are compatible with the phrase query being evaluated. This requires working out offsets between the words.



Example 2.1: Satisfying phrase queries. Suppose the postings lists for *to* and *be* are as in Figure 2.11, and the query is “*to be or not to be*”. The postings lists to access are: *to*, *be*, or, not. We will examine intersecting the postings lists for *to* and *be*. We first look for documents that contain both terms. Then, we look for places in the lists where there is an occurrence of *be* with a token index one higher than a position of *to*, and then we look for another occurrence of each word with token index 4 higher than the first occurrence. In the above lists, the pattern of occurrences that is a possible match is:

```

to: ⟨...; 4:⟨..., 429, 433⟩; ...⟩
be: ⟨...; 4:⟨..., 430, 434⟩; ...⟩

```

```

POSITIONALINTERSECT( $p_1, p_2, k$ )
1 answer  $\leftarrow \langle \rangle$ 
2 while  $p_1 \neq \text{NIL}$  and  $p_2 \neq \text{NIL}$ 
3 do if  $\text{docID}(p_1) = \text{docID}(p_2)$ 
4   then  $l \leftarrow \langle \rangle$ 
5      $pp_1 \leftarrow \text{positions}(p_1)$ 
6      $pp_2 \leftarrow \text{positions}(p_2)$ 
7     while  $pp_1 \neq \text{NIL}$ 
8       do while  $pp_2 \neq \text{NIL}$ 
9         do if  $|\text{pos}(pp_1) - \text{pos}(pp_2)| \leq k$ 
10        then ADD( $l, \text{pos}(pp_2)$ )
11        else if  $\text{pos}(pp_2) > \text{pos}(pp_1)$ 
12          then break
13         $pp_2 \leftarrow \text{next}(pp_2)$ 
14        while  $l \neq \langle \rangle$  and  $|l[0] - \text{pos}(pp_1)| > k$ 
15          do DELETE( $l[0]$ )
16          for each  $ps \in l$ 
17            do ADD(answer,  $\langle \text{docID}(p_1), \text{pos}(pp_1), ps \rangle$ )
18             $pp_1 \leftarrow \text{next}(pp_1)$ 
19             $p_1 \leftarrow \text{next}(p_1)$ 
20             $p_2 \leftarrow \text{next}(p_2)$ 
21          else if  $\text{docID}(p_1) < \text{docID}(p_2)$ 
22            then  $p_1 \leftarrow \text{next}(p_1)$ 
23            else  $p_2 \leftarrow \text{next}(p_2)$ 
24 return answer

```

► **Figure 2.12** An algorithm for proximity intersection of postings lists p_1 and p_2 . The algorithm finds places where the two terms appear within k words of each other and returns a list of triples giving docID and the term position in p_1 and p_2 .

The same general method is applied for within k word proximity searches, of the sort we saw in Example 1.1 (page 15):

employment /3 place

Here, $/k$ means “within k words of (on either side)”. Clearly, positional indexes can be used for such queries; biword indexes cannot. We show in Figure 2.12 an algorithm for satisfying within k word proximity searches; it is further discussed in Exercise 2.12.

Positional index size. Adopting a positional index expands required postings storage significantly, even if we compress position values/offsets as we

will discuss in Section 5.3 (page 95). Indeed, moving to a positional index also changes the asymptotic complexity of a postings intersection operation, because the number of items to check is now bounded not by the number of documents but by the total number of tokens in the document collection T . That is, the complexity of a Boolean query is $\Theta(T)$ rather than $\Theta(N)$. However, most applications have little choice but to accept this, since most users now expect to have the functionality of phrase and proximity searches.

Let's examine the space implications of having a positional index. A posting now needs an entry for each occurrence of a term. The index size thus depends on the average document size. The average web page has less than 1000 terms, but documents like SEC stock filings, books, and even some epic poems easily reach 100,000 terms. Consider a term with frequency 1 in 1000 terms on average. The result is that large documents cause an increase of two orders of magnitude in the space required to store the postings list:

Document size	Expected postings	Expected entries in positional posting
1000	1	1
100,000	1	100

While the exact numbers depend on the type of documents and the language being indexed, some rough rules of thumb are to expect a positional index to be 2 to 4 times as large as a non-positional index, and to expect a compressed positional index to be about one third to one half the size of the raw text (after removal of markup, etc.) of the original uncompressed documents. Specific numbers for an example collection are given in Table 5.1 (page 87) and Table 5.6 (page 103).

2.4.3 Combination schemes

The strategies of biword indexes and positional indexes can be fruitfully combined. If users commonly query on particular phrases, such as Michael Jackson, it is quite inefficient to keep merging positional postings lists. A combination strategy uses a phrase index, or just a biword index, for certain queries and uses a positional index for other phrase queries. Good queries to include in the phrase index are ones known to be common based on recent querying behavior. But this is not the only criterion: the most expensive phrase queries to evaluate are ones where the individual words are common but the desired phrase is comparatively rare. Adding *Britney Spears* as a phrase index entry may only give a speedup factor to that query of about 3, since most documents that mention either word are valid results, whereas adding *The Who* as a phrase index entry may speed up that query by a factor of 1000. Hence, having the latter is more desirable, even if it is a relatively less common query.

NEXT WORD INDEX

Williams et al. (2004) evaluate an even more sophisticated scheme which employs indexes of both these sorts and additionally a partial next word index as a halfway house between the first two strategies. For each term, a *next word index* records terms that follow it in a document. They conclude that such a strategy allows a typical mixture of web phrase queries to be completed in one quarter of the time taken by use of a positional index alone, while taking up 26% more space than use of a positional index alone.

**Exercise 2.8**

[★]

Assume a biword index. Give an example of a document which will be returned for a query of New York University but is actually a false positive which should not be returned.

Exercise 2.9

[★]

Shown below is a portion of a positional index in the format: term: doc1: ⟨position1, position2, ...⟩; doc2: ⟨position1, position2, ...⟩; etc.

```

angels: 2: ⟨36,174,252,651⟩; 4: ⟨12,22,102,432⟩; 7: ⟨17⟩;
fools: 2: ⟨1,17,74,222⟩; 4: ⟨8,78,108,458⟩; 7: ⟨3,13,23,193⟩;
fear: 2: ⟨87,704,722,901⟩; 4: ⟨13,43,113,433⟩; 7: ⟨18,328,528⟩;
in: 2: ⟨3,37,76,444,851⟩; 4: ⟨10,20,110,470,500⟩; 7: ⟨5,15,25,195⟩;
rush: 2: ⟨2,66,194,321,702⟩; 4: ⟨9,69,149,429,569⟩; 7: ⟨4,14,404⟩;
to: 2: ⟨47,86,234,999⟩; 4: ⟨14,24,774,944⟩; 7: ⟨199,319,599,709⟩;
tread: 2: ⟨57,94,333⟩; 4: ⟨15,35,155⟩; 7: ⟨20,320⟩;
where: 2: ⟨67,124,393,1001⟩; 4: ⟨11,41,101,421,431⟩; 7: ⟨16,36,736⟩;
```

Which document(s) if any match each of the following queries, where each expression within quotes is a phrase query?

- “fools rush in”
- “fools rush in” AND “angels fear to tread”

Exercise 2.10

[★]

Consider the following fragment of a positional index with the format:

```

word: document: ⟨position, position, ...⟩; document: ⟨position, ...⟩
...
```

```

Gates: 1: ⟨3⟩; 2: ⟨6⟩; 3: ⟨2,17⟩; 4: ⟨1⟩;
IBM: 4: ⟨3⟩; 7: ⟨14⟩;
Microsoft: 1: ⟨1⟩; 2: ⟨1,21⟩; 3: ⟨3⟩; 5: ⟨16,22,51⟩;
```

The $/k$ operator, $\text{word1} /k \text{word2}$ finds occurrences of word1 within k words of word2 (on either side), where k is a positive integer argument. Thus $k = 1$ demands that word1 be adjacent to word2 .

- Describe the set of documents that satisfy the query Gates /2 Microsoft.
- Describe each set of values for k for which the query Gates / k Microsoft returns a different set of documents as the answer.

Exercise 2.11

[**]

Consider the general procedure for merging two positional postings lists for a given document, to determine the document positions where a document satisfies a $/k$ clause (in general there can be multiple positions at which each term occurs in a single document). We begin with a pointer to the position of occurrence of each term and move each pointer along the list of occurrences in the document, checking as we do so whether we have a hit for $/k$. Each move of either pointer counts as a step. Let L denote the total number of occurrences of the two terms in the document. What is the big-O complexity of the merge procedure, if we wish to have postings including positions in the result?

Exercise 2.12

[**]

Consider the adaptation of the basic algorithm for intersection of two postings lists (Figure 1.6, page 11) to the one in Figure 2.12 (page 42), which handles proximity queries. A naive algorithm for this operation could be $O(PL_{\max}^2)$, where P is the sum of the lengths of the postings lists (i.e., the sum of document frequencies) and L_{\max} is the maximum length of a document (in tokens).

- Go through this algorithm carefully and explain how it works.
- What is the complexity of this algorithm? Justify your answer carefully.
- For certain queries and data distributions, would another algorithm be more efficient? What complexity does it have?

Exercise 2.13

[**]

Suppose we wish to use a postings intersection procedure to determine simply the list of documents that satisfy a $/k$ clause, rather than returning the list of positions, as in Figure 2.12 (page 42). For simplicity, assume $k \geq 2$. Let L denote the total number of occurrences of the two terms in the document collection (i.e., the sum of their collection frequencies). Which of the following is true? Justify your answer.

- The merge can be accomplished in a number of steps linear in L and independent of k , and we can ensure that each pointer moves only to the right.
- The merge can be accomplished in a number of steps linear in L and independent of k , but a pointer may be forced to move non-monotonically (i.e., to sometimes back up)
- The merge can require kL steps in some cases.

Exercise 2.14

[**]

How could an IR system combine use of a positional index and use of stop words? What is the potential problem, and how could it be handled?

2.5 References and further reading

EAST ASIAN
LANGUAGES

Exhaustive discussion of the character-level processing of East Asian languages can be found in Lunde (1998). Character bigram indexes are perhaps the most standard approach to indexing Chinese, although some systems use word segmentation. Due to differences in the language and writing system, word segmentation is most usual for Japanese (Luk and Kwok 2002, Kishida

[et al.](#) 2005). The structure of a character k -gram index over unsegmented text differs from that in Section 3.2.2 (page 54): there the k -gram dictionary points to postings lists of entries in the regular dictionary, whereas here it points directly to document postings lists. For further discussion of Chinese word segmentation, see [Sproat et al. \(1996\)](#), [Sproat and Emerson \(2003\)](#), [Tseng et al. \(2005\)](#), and [Gao et al. \(2005\)](#).

[Lita et al. \(2003\)](#) present a method for truecasing. Natural language processing work on computational morphology is presented in ([Sproat 1992](#), [Beesley and Karttunen 2003](#)).

Language identification was perhaps first explored in cryptography; for example, [Konheim \(1981\)](#) presents a character-level k -gram language identification algorithm. While other methods such as looking for particular distinctive function words and letter combinations have been used, with the advent of widespread digital text, many people have explored the character n -gram technique, and found it to be highly successful ([Beesley 1998](#), [Dunning 1994](#), [Cavnar and Trenkle 1994](#)). Written language identification is regarded as a fairly easy problem, while spoken language identification remains more difficult; see [Hughes et al. \(2006\)](#) for a recent survey.

Experiments on and discussion of the positive and negative impact of stemming in English can be found in the following works: [Salton \(1989\)](#), [Harman \(1991\)](#), [Krovetz \(1995\)](#), [Hull \(1996\)](#). [Hollink et al. \(2004\)](#) provide detailed results for the effectiveness of language-specific methods on 8 European languages. In terms of percent change in mean average precision (see page 159) over a baseline system, diacritic removal gains up to 23% (being especially helpful for Finnish, French, and Swedish). Stemming helped markedly for Finnish (30% improvement) and Spanish (10% improvement), but for most languages, including English, the gain from stemming was in the range 0–5%, and results from a lemmatizer were poorer still. Compound splitting gained 25% for Swedish and 15% for German, but only 4% for Dutch. Rather than language-particular methods, indexing character k -grams (as we suggested for Chinese) could often give as good or better results: using within-word character 4-grams rather than words gave gains of 37% in Finnish, 27% in Swedish, and 20% in German, while even being slightly positive for other languages, such as Dutch, Spanish, and English. [Tomlinson \(2003\)](#) presents broadly similar results. [Bar-Ilan and Gutman \(2005\)](#) suggest that, at the time of their study (2003), the major commercial web search engines suffered from lacking decent language-particular processing; for example, a query on www.google.fr for l'électricité did not separate off the article l' but only matched pages with precisely this string of article+noun.

SKIP LIST

The classic presentation of skip pointers for IR can be found in [Moffat and Zobel \(1996\)](#). Extended techniques are discussed in [Boldi and Vigna \(2005\)](#). The main paper in the algorithms literature is [Pugh \(1990\)](#), which uses multilevel skip pointers to give expected $O(\log P)$ list access (the same expected

efficiency as using a tree data structure) with less implementational complexity. In practice, the effectiveness of using skip pointers depends on various system parameters. [Moffat and Zobel \(1996\)](#) report conjunctive queries running about five times faster with the use of skip pointers, but [Bahle et al. \(2002, p. 217\)](#) report that, with modern CPUs, using skip lists instead slows down search because it expands the size of the postings list (i.e., disk I/O dominates performance). In contrast, [Strohman and Croft \(2007\)](#) again show good performance gains from skipping, in a system architecture designed to optimize for the large memory spaces and multiple cores of recent CPUs.

[Johnson et al. \(2006\)](#) report that 11.7% of all queries in two 2002 web query logs contained phrase queries, though [Kammenhuber et al. \(2006\)](#) report only 3% phrase queries for a different data set. [Silverstein et al. \(1999\)](#) note that many queries without explicit phrase operators are actually implicit phrase searches.

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3 *Dictionaries and tolerant retrieval*

WILDCARD QUERY

In Chapters 1 and 2 we developed the ideas underlying inverted indexes for handling Boolean and proximity queries. Here, we develop techniques that are robust to typographical errors in the query, as well as alternative spellings. In Section 3.1 we develop data structures that help the search for terms in the vocabulary in an inverted index. In Section 3.2 we study the idea of a *wildcard query*: a query such as **a*e*i*o*u**, which seeks documents containing any term that includes all the five vowels in sequence. The *** symbol indicates any (possibly empty) string of characters. Users pose such queries to a search engine when they are uncertain about how to spell a query term, or seek documents containing variants of a query term; for instance, the query *automat** would seek documents containing any of the terms *automatic*, *automation* and *automated*.

We then turn to other forms of imprecisely posed queries, focusing on spelling errors in Section 3.3. Users make spelling errors either by accident, or because the term they are searching for (e.g., *Herman*) has no unambiguous spelling in the collection. We detail a number of techniques for correcting spelling errors in queries, one term at a time as well as for an entire string of query terms. Finally, in Section 3.4 we study a method for seeking vocabulary terms that are phonetically close to the query term(s). This can be especially useful in cases like the *Herman* example, where the user may not know how a proper name is spelled in documents in the collection.

Because we will develop many variants of inverted indexes in this chapter, we will use sometimes the phrase *standard inverted index* to mean the inverted index developed in Chapters 1 and 2, in which each vocabulary term has a postings list with the documents in the collection.

3.1 Search structures for dictionaries

Given an inverted index and a query, our first task is to determine whether each query term exists in the vocabulary and if so, identify the pointer to the

corresponding postings. This vocabulary lookup operation uses a classical data structure called the dictionary and has two broad classes of solutions: hashing, and search trees. In the literature of data structures, the entries in the vocabulary (in our case, terms) are often referred to as *keys*. The choice of solution (hashing, or search trees) is governed by a number of questions: (1) How many keys are we likely to have? (2) Is the number likely to remain static, or change a lot – and in the case of changes, are we likely to only have new keys inserted, or to also have some keys in the dictionary be deleted? (3) What are the relative frequencies with which various keys will be accessed?

Hashing has been used for dictionary lookup in some search engines. Each vocabulary term (key) is hashed into an integer over a large enough space that hash collisions are unlikely; collisions if any are resolved by auxiliary structures that can demand care to maintain.¹ At query time, we hash each query term separately and following a pointer to the corresponding postings, taking into account any logic for resolving hash collisions. There is no easy way to find minor variants of a query term (such as the accented and non-accented versions of a word like *resume*), since these could be hashed to very different integers. In particular, we cannot seek (for instance) all terms beginning with the prefix *automat*, an operation that we will require below in Section 3.2. Finally, in a setting (such as the Web) where the size of the vocabulary keeps growing, a hash function designed for current needs may not suffice in a few years' time.

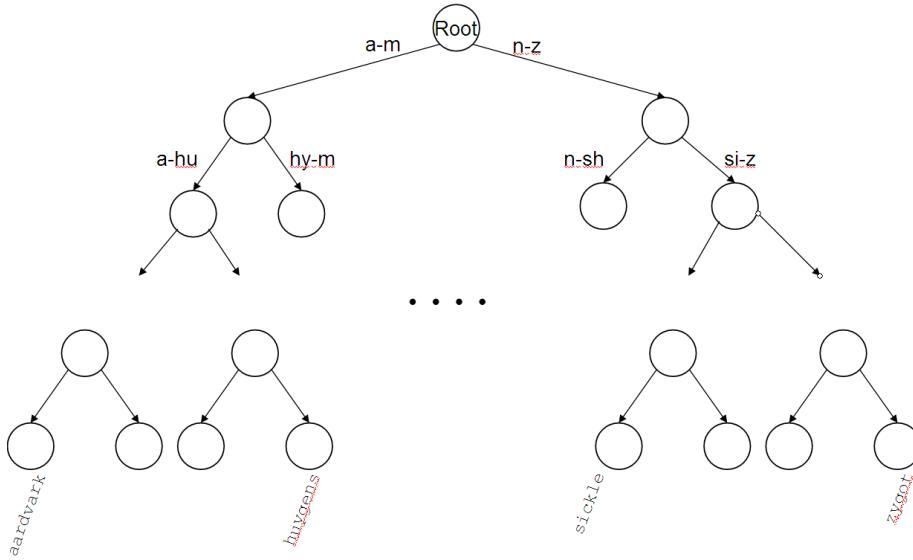
BINARY TREE

Search trees overcome many of these issues – for instance, they permit us to enumerate all vocabulary terms beginning with *automat*. The best-known search tree is the *binary tree*, in which each internal node has two children. The search for a term begins at the root of the tree. Each internal node (including the root) represents a binary test, based on whose outcome the search proceeds to one of the two sub-trees below that node. Figure 3.1 gives an example of a binary search tree used for a dictionary. Efficient search (with a number of comparisons that is $O(\log M)$) hinges on the tree being balanced: the numbers of terms under the two sub-trees of any node are either equal or differ by one. The principal issue here is that of rebalancing: as terms are inserted into or deleted from the binary search tree, it needs to be rebalanced so that the balance property is maintained.

B-TREE

To mitigate rebalancing, one approach is to allow the number of sub-trees under an internal node to vary in a fixed interval. A search tree commonly used for a dictionary is the *B-tree* – a search tree in which every internal node has a number of children in the interval $[a, b]$, where a and b are appropriate positive integers; Figure 3.2 shows an example with $a = 2$ and $b = 4$. Each branch under an internal node again represents a test for a range of char-

¹ So-called perfect hash functions are designed to preclude collisions, but are rather more complicated both to implement and to compute.



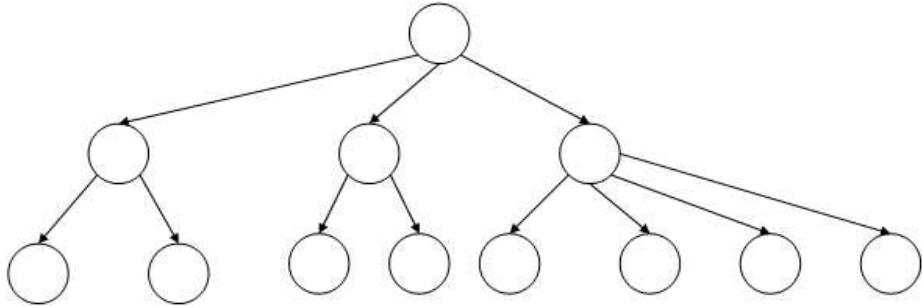
► **Figure 3.1** A binary search tree. In this example the branch at the root partitions vocabulary terms into two subtrees, those whose first letter is between a and m, and the rest.

acter sequences, as in the binary tree example of Figure 3.1. A B-tree may be viewed as “collapsing” multiple levels of the binary tree into one; this is especially advantageous when some of the dictionary is disk-resident, in which case this collapsing serves the function of pre-fetching imminent binary tests. In such cases, the integers a and b are determined by the sizes of disk blocks. Section 3.5 contains pointers to further background on search trees and B-trees.

It should be noted that unlike hashing, search trees demand that the characters used in the document collection have a prescribed ordering; for instance, the 26 letters of the English alphabet are always listed in the specific order A through Z. Some Asian languages such as Chinese do not always have a unique ordering, although by now all languages (including Chinese and Japanese) have adopted a standard ordering system for their character sets.

3.2 Wildcard queries

Wildcard queries are used in any of the following situations: (1) the user is uncertain of the spelling of a query term (e.g., Sydney vs. Sidney, which



► **Figure 3.2** A B-tree. In this example every internal node has between 2 and 4 children.

leads to the wildcard query S^*dney); (2) the user is aware of multiple variants of spelling a term and (consciously) seeks documents containing any of the variants (e.g., color vs. colour); (3) the user seeks documents containing variants of a term that would be caught by stemming, but is unsure whether the search engine performs stemming (e.g., judicial vs. judiciary, leading to the wildcard query $judicia^*$); (4) the user is uncertain of the correct rendition of a foreign word or phrase (e.g., the query $Universit^*$ Stuttgart).

WILDCARD QUERY

A query such as mon^* is known as a *trailing wildcard query*, because the $*$ symbol occurs only once, at the end of the search string. A search tree on the dictionary is a convenient way of handling trailing wildcard queries: we walk down the tree following the symbols m, o and n in turn, at which point we can enumerate the set W of terms in the dictionary with the prefix mon. Finally, we use $|W|$ lookups on the standard inverted index to retrieve all documents containing any term in W .

But what about wildcard queries in which the $*$ symbol is not constrained to be at the end of the search string? Before handling this general case, we mention a slight generalization of trailing wildcard queries. First, consider *leading wildcard queries*, or queries of the form *mon . Consider a *reverse B-tree* on the dictionary – one in which each root-to-leaf path of the B-tree corresponds to a term in the dictionary written *backwards*: thus, the term lemon would, in the B-tree, be represented by the path root-n-o-m-e-l. A walk down the reverse B-tree then enumerates all terms R in the vocabulary with a given prefix.

In fact, using a regular B-tree together with a reverse B-tree, we can handle an even more general case: wildcard queries in which there is a single $*$ symbol, such as se^*mon . To do this, we use the regular B-tree to enumerate the set W of dictionary terms beginning with the prefix se, then the reverse B-tree to

enumerate the set R of terms ending with the suffix mon. Next, we take the intersection $W \cap R$ of these two sets, to arrive at the set of terms that begin with the prefix se and end with the suffix mon. Finally, we use the standard inverted index to retrieve all documents containing any terms in this intersection. We can thus handle wildcard queries that contain a single * symbol using two B-trees, the normal B-tree and a reverse B-tree.

3.2.1 General wildcard queries

We now study two techniques for handling general wildcard queries. Both techniques share a common strategy: express the given wildcard query q_w as a Boolean query Q on a specially constructed index, such that the answer to Q is a superset of the set of vocabulary terms matching q_w . Then, we check each term in the answer to Q against q_w , discarding those vocabulary terms that do not match q_w . At this point we have the vocabulary terms matching q_w and can resort to the standard inverted index.

Permuterm indexes

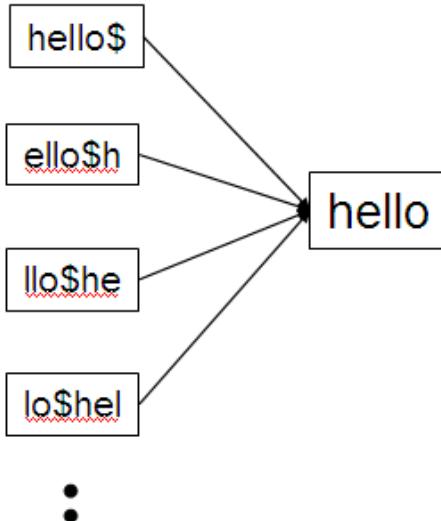
PERMUTERM INDEX

Our first special index for general wildcard queries is the *permuterm index*, a form of inverted index. First, we introduce a special symbol \$ into our character set, to mark the end of a term. Thus, the term hello is shown here as the augmented term hello\$. Next, we construct a permuterm index, in which the various rotations of each term (augmented with \$) all link to the original vocabulary term. Figure 3.3 gives an example of such a permuterm index entry for the term hello.

We refer to the set of rotated terms in the permuterm index as the *permuterm vocabulary*.

How does this index help us with wildcard queries? Consider the wildcard query m*n. The key is to *rotate* such a wildcard query so that the * symbol appears at the end of the string – thus the rotated wildcard query becomes n\$m*. Next, we look up this string in the permuterm index, where seeking n\$m* (via a search tree) leads to rotations of (among others) the terms man and moron.

Now that the permuterm index enables us to identify the original vocabulary terms matching a wildcard query, we look up these terms in the standard inverted index to retrieve matching documents. We can thus handle any wildcard query with a single * symbol. But what about a query such as fi*mo*er? In this case we first enumerate the terms in the dictionary that are in the permuterm index of er\$fi*. Not all such dictionary terms will have the string mo in the middle - we filter these out by exhaustive enumeration, checking each candidate to see if it contains mo. In this example, the term fishmonger would survive this filtering but filibuster would not. We then



► Figure 3.3 A portion of a permuterm index.

run the surviving terms through the standard inverted index for document retrieval. One disadvantage of the permuterm index is that its dictionary becomes quite large, including as it does all rotations of each term.

Notice the close interplay between the B-tree and the permuterm index above. Indeed, it suggests that the structure should perhaps be viewed as a permuterm B-tree. However, we follow traditional terminology here in describing the permuterm index as distinct from the B-tree that allows us to select the rotations with a given prefix.

3.2.2 *k*-gram indexes for wildcard queries

Whereas the permuterm index is simple, it can lead to a considerable blowup from the number of rotations per term; for a dictionary of English terms, this can represent an almost ten-fold space increase. We now present a second technique, known as the *k*-gram index, for processing wildcard queries. We will also use *k*-gram indexes in Section 3.3.4. A *k*-gram is a sequence of *k* characters. Thus cas, ast and stl are all 3-grams occurring in the term castle. We use a special character \$ to denote the beginning or end of a term, so the full set of 3-grams generated for castle is: \$ca, ca\$, ast, stl, tle, le\$.

k-GRAM INDEX

In a *k*-gram index, the dictionary contains all *k*-grams that occur in any term in the vocabulary. Each postings list points from a *k*-gram to all vocabulary



► **Figure 3.4** Example of a postings list in a 3-gram index. Here the 3-gram *etr* is illustrated. Matching vocabulary terms are lexicographically ordered in the postings.

terms containing that k -gram. For instance, the 3-gram *etr* would point to vocabulary terms such as *metric* and *retrieval*. An example is given in Figure 3.4.

How does such an index help us with wildcard queries? Consider the wildcard query *re*ve*. We are seeking documents containing any term that begins with *re* and ends with *ve*. Accordingly, we run the Boolean query `$re AND ve$`. This is looked up in the 3-gram index and yields a list of matching terms such as *relive*, *remove* and *retrieve*. Each of these matching terms is then looked up in the standard inverted index to yield documents matching the query.

There is however a difficulty with the use of k -gram indexes, that demands one further step of processing. Consider using the 3-gram index described above for the query *red**. Following the process described above, we first issue the Boolean query `$re AND red` to the 3-gram index. This leads to a match on terms such as *retired*, which contain the conjunction of the two 3-grams *\$re* and *red*, yet do not match the original wildcard query *red**.

To cope with this, we introduce a *post-filtering* step, in which the terms enumerated by the Boolean query on the 3-gram index are checked individually against the original query *red**. This is a simple string-matching operation and weeds out terms such as *retired* that do not match the original query. Terms that survive are then searched in the standard inverted index as usual.

We have seen that a wildcard query can result in multiple terms being enumerated, each of which becomes a single-term query on the standard inverted index. Search engines do allow the combination of wildcard queries using Boolean operators, for example, *re*d AND fe*ri*. What is the appropriate semantics for such a query? Since each wildcard query turns into a disjunction of single-term queries, the appropriate interpretation of this example is that we have a conjunction of disjunctions: we seek all documents that contain any term matching *re*d* *and* any term matching *fe*ri*.

Even without Boolean combinations of wildcard queries, the processing of a wildcard query can be quite expensive, because of the added lookup in the special index, filtering and finally the standard inverted index. A search engine may support such rich functionality, but most commonly, the capability is hidden behind an interface (say an “Advanced Query” interface) that most

users never use. Exposing such functionality in the search interface often encourages users to invoke it even when they do not require it (say, by typing a prefix of their query followed by a *), increasing the processing load on the search engine.



Exercise 3.1

In the permuterm index, each permuterm vocabulary term points to the original vocabulary term(s) from which it was derived. How many original vocabulary terms can there be in the postings list of a permuterm vocabulary term?

Exercise 3.2

Write down the entries in the permuterm index dictionary that are generated by the term mama.

Exercise 3.3

If you wanted to search for s*ng in a permuterm wildcard index, what key(s) would one do the lookup on?

Exercise 3.4

Refer to Figure 3.4; it is pointed out in the caption that the vocabulary terms in the postings are lexicographically ordered. Why is this ordering useful?

Exercise 3.5

Consider again the query fi*mo*er from Section 3.2.1. What Boolean query on a bigram index would be generated for this query? Can you think of a term that matches the permuterm query in Section 3.2.1, but does not satisfy this Boolean query?

Exercise 3.6

Give an example of a sentence that falsely matches the wildcard query mon*h if the search were to simply use a conjunction of bigrams.

3.3 Spelling correction

We next look at the problem of correcting spelling errors in queries. For instance, we may wish to retrieve documents containing the term carrot when the user types the query carot. Google reports (<http://www.google.com/jobs/britney.html>) that the following are all treated as misspellings of the query britney spears: britian spears, britney's spears, brandy spears and prittany spears. We look at two steps to solving this problem: the first based on *edit distance* and the second based on *k-gram overlap*. Before getting into the algorithmic details of these methods, we first review how search engines provide spell-correction as part of a user experience.

3.3.1 Implementing spelling correction

There are two basic principles underlying most spelling correction algorithms.

1. Of various alternative correct spellings for a mis-spelled query, choose the “nearest” one. This demands that we have a notion of nearness or proximity between a pair of queries. We will develop these proximity measures in Section 3.3.3.
2. When two correctly spelled queries are tied (or nearly tied), select the one that is more common. For instance, *grunt* and *grant* both seem equally plausible as corrections for *grnt*. Then, the algorithm should choose the more common of *grunt* and *grant* as the correction. The simplest notion of more common is to consider the number of occurrences of the term in the collection; thus if *grunt* occurs more often than *grant*, it would be the chosen correction. A different notion of more common is employed in many search engines, especially on the web. The idea is to use the correction that is most common among queries typed in by other users. The idea here is that if *grunt* is typed as a query more often than *grant*, then it is more likely that the user who typed *grnt* intended to type the query *grunt*.

Beginning in Section 3.3.3 we describe notions of proximity between queries, as well as their efficient computation. Spelling correction algorithms build on these computations of proximity; their functionality is then exposed to users in one of several ways:

1. On the query *carot* always retrieve documents containing *carot* as well as any “spell-corrected” version of *carot*, including *carrot* and *tarot*.
2. As in (1) above, but only when the query term *carot* is not in the dictionary.
3. As in (1) above, but only when the original query returned fewer than a preset number of documents (say fewer than five documents).
4. When the original query returns fewer than a preset number of documents, the search interface presents a *spelling suggestion* to the end user: this suggestion consists of the spell-corrected query term(s). Thus, the search engine might respond to the user: “Did you mean *carrot*? ”

3.3.2 Forms of spelling correction

We focus on two specific forms of spelling correction that we refer to as *isolated-term* correction and *context-sensitive* correction. In isolated-term correction, we attempt to correct a single query term at a time – even when we

have a multiple-term query. The carot example demonstrates this type of correction. Such isolated-term correction would fail to detect, for instance, that the query flew form Heathrow contains a mis-spelling of the term from – because each term in the query is correctly spelled in isolation.

We begin by examining two techniques for addressing isolated-term correction: edit distance, and k -gram overlap. We then proceed to context-sensitive correction.

3.3.3 Edit distance

EDIT DISTANCE

Given two character strings s_1 and s_2 , the *edit distance* between them is the minimum number of *edit operations* required to transform s_1 into s_2 . Most commonly, the edit operations allowed for this purpose are: (i) insert a character into a string; (ii) delete a character from a string and (iii) replace a character of a string by another character; for these operations, edit distance is sometimes known as *Levenshtein distance*. For example, the edit distance between cat and dog is 3. In fact, the notion of edit distance can be generalized to allowing different weights for different kinds of edit operations, for instance a higher weight may be placed on replacing the character s by the character p, than on replacing it by the character a (the latter being closer to s on the keyboard). Setting weights in this way depending on the likelihood of letters substituting for each other is very effective in practice (see Section 3.4 for the separate issue of phonetic similarity). However, the remainder of our treatment here will focus on the case in which all edit operations have the same weight.

It is well-known how to compute the (weighted) edit distance between two strings in time $O(|s_1| \times |s_2|)$, where $|s_i|$ denotes the length of a string s_i . The idea is to use the dynamic programming algorithm in Figure 3.5, where the characters in s_1 and s_2 are given in array form. The algorithm fills the (integer) entries in a matrix m whose two dimensions equal the lengths of the two strings whose edit distances is being computed; the (i, j) entry of the matrix will hold (after the algorithm is executed) the edit distance between the strings consisting of the first i characters of s_1 and the first j characters of s_2 . The central dynamic programming step is depicted in Lines 8-10 of Figure 3.5, where the three quantities whose minimum is taken correspond to substituting a character in s_1 , inserting a character in s_1 and inserting a character in s_2 .

Figure 3.6 shows an example Levenshtein distance computation of Figure 3.5. The typical cell $[i, j]$ has four entries formatted as a 2×2 cell. The lower right entry in each cell is the min of the other three, corresponding to the main dynamic programming step in Figure 3.5. The other three entries are the three entries $m[i - 1, j - 1] + 0$ or 1 depending on whether $s_1[i] =$

```

EDITDISTANCE( $s_1, s_2$ )
1  int  $m[i, j] = 0$ 
2  for  $i \leftarrow 1$  to  $|s_1|$ 
3  do  $m[i, 0] = i$ 
4  for  $j \leftarrow 1$  to  $|s_2|$ 
5  do  $m[0, j] = j$ 
6  for  $i \leftarrow 1$  to  $|s_1|$ 
7  do for  $j \leftarrow 1$  to  $|s_2|$ 
8    do  $m[i, j] = \min\{m[i - 1, j - 1] + \text{if } (s_1[i] = s_2[j]) \text{ then } 0 \text{ else } 1, m[i - 1, j] + 1,$ 
9       $m[i, j - 1] + 1\}$ 
10
11 return  $m[|s_1|, |s_2|]$ 

```

► **Figure 3.5** Dynamic programming algorithm for computing the edit distance between strings s_1 and s_2 .

		f	a	s	t
	0	1 1	2 2	3 3	4 4
c	1 1	1 2 2 1	2 3 2 2	3 4 3 3	4 5 4 4
a	2 2	2 2 3 2	1 3 3 1	3 4 2 2	4 5 3 3
t	3 3	3 3 4 3	3 2 4 2	2 3 3 2	2 4 3 2
s	4 4	4 4 5 4	4 3 5 3	2 3 4 2	3 3 3 3

► **Figure 3.6** Example Levenshtein distance computation. The 2×2 cell in the $[i, j]$ entry of the table shows the three numbers whose minimum yields the fourth. The cells in italics determine the edit distance in this example.

$s_2[j]$, $m[i - 1, j] + 1$ and $m[i, j - 1] + 1$. The cells with numbers in italics depict the path by which we determine the Levenshtein distance.

The spelling correction problem however demands more than computing edit distance: given a set S of strings (corresponding to terms in the vocabulary) and a query string q , we seek the string(s) in V of least edit distance from q . We may view this as a decoding problem, in which the codewords (the strings in V) are prescribed in advance. The obvious way of doing this is to compute the edit distance from q to each string in V , before selecting the

string(s) of minimum edit distance. This exhaustive search is inordinately expensive. Accordingly, a number of heuristics are used in practice to efficiently retrieve vocabulary terms likely to have low edit distance to the query term(s).

The simplest such heuristic is to restrict the search to dictionary terms beginning with the same letter as the query string; the hope would be that spelling errors do not occur in the first character of the query. A more sophisticated variant of this heuristic is to use a version of the permuterm index, in which we omit the end-of-word symbol \$. Consider the set of all rotations of the query string q . For each rotation r from this set, we traverse the B-tree into the permuterm index, thereby retrieving all dictionary terms that have a rotation beginning with r . For instance, if q is mase and we consider the rotation $r = \text{sema}$, we would retrieve dictionary terms such as semantic and semaphore that do not have a small edit distance to q . Unfortunately, we would miss more pertinent dictionary terms such as mare and mane. To address this, we refine this rotation scheme: for each rotation, we omit a suffix of ℓ characters before performing the B-tree traversal. This ensures that each term in the set R of terms retrieved from the dictionary includes a “long” substring in common with q . The value of ℓ could depend on the length of q . Alternatively, we may set it to a fixed constant such as 2.

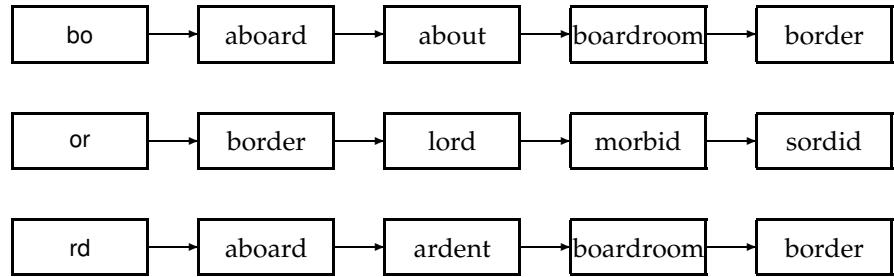
3.3.4 k -gram indexes for spelling correction

To further limit the set of vocabulary terms for which we compute edit distances to the query term, we now show how to invoke the k -gram index of Section 3.2.2 (page 54) to assist with retrieving vocabulary terms with low edit distance to the query q . Once we retrieve such terms, we can then find the ones of least edit distance from q .

In fact, we will use the k -gram index to retrieve vocabulary terms that have many k -grams in common with the query. We will argue that for reasonable definitions of “many k -grams in common,” the retrieval process is essentially that of a single scan through the postings for the k -grams in the query string q .

The 2-gram (or *bigram*) index in Figure 3.7 shows (a portion of) the postings for the three bigrams in the query bord. Suppose we wanted to retrieve vocabulary terms that contained at least two of these three bigrams. A single scan of the postings (much as in Chapter 1) would let us enumerate all such terms; in the example of Figure 3.7 we would enumerate aboard, boardroom and border.

This straightforward application of the linear scan intersection of postings immediately reveals the shortcoming of simply requiring matched vocabulary terms to contain a fixed number of k -grams from the query q : terms like boardroom, an implausible “correction” of bord, get enumerated. Conse-



► **Figure 3.7** Matching at least two of the three 2-grams in the query bord.

JACCARD COEFFICIENT

quently, we require more nuanced measures of the overlap in k -grams between a vocabulary term and q . The linear scan intersection can be adapted when the measure of overlap is the *Jaccard coefficient* for measuring the overlap between two sets A and B , defined to be $|A \cap B| / |A \cup B|$. The two sets we consider are the set of k -grams in the query q , and the set of k -grams in a vocabulary term. As the scan proceeds, we proceed from one vocabulary term t to the next, computing on the fly the Jaccard coefficient between q and t . If the coefficient exceeds a preset threshold, we add t to the output; if not, we move on to the next term in the postings. To compute the Jaccard coefficient, we need the set of k -grams in q and t .

Since we are scanning the postings for all k -grams in q , we immediately have these k -grams on hand. What about the k -grams of t ? In principle, we could enumerate these on the fly from t ; in practice this is not only slow but potentially infeasible since, in all likelihood, the postings entries themselves do not contain the complete string t but rather some encoding of t . The crucial observation is that to compute the Jaccard coefficient, we only need the length of the string t . To see this, recall the example of Figure 3.7 and consider the point when the postings scan for query $q = \text{bord}$ reaches term $t = \text{boardroom}$. We know that two bigrams match. If the postings stored the (pre-computed) number of bigrams in boardroom (namely, 8), we have all the information we require to compute the Jaccard coefficient to be $2 / (8 + 3 - 2)$; the numerator is obtained from the number of postings hits (2, from bo and rd) while the denominator is the sum of the number of bigrams in bord and boardroom , less the number of postings hits.

We could replace the Jaccard coefficient by other measures that allow efficient on the fly computation during postings scans. How do we use these

for spelling correction? One method that has some empirical support is to first use the k -gram index to enumerate a set of candidate vocabulary terms that are potential corrections of q . We then compute the edit distance from q to each term in this set, selecting terms from the set with small edit distance to q .

3.3.5 Context sensitive spelling correction

Isolated-term correction would fail to correct typographical errors such as *flew* form *Heathrow*, where all three query terms are correctly spelled. When a phrase such as this retrieves few documents, a search engine may like to offer the corrected query *flew* from *Heathrow*. The simplest way to do this is to enumerate corrections of each of the three query terms (using the methods leading up to Section 3.3.4) even though each query term is correctly spelled, then try substitutions of each correction in the phrase. For the example *flew* from *Heathrow*, we enumerate such phrases as *fled* from *Heathrow* and *flew* fore *Heathrow*. For each such substitute phrase, the search engine runs the query and determines the number of matching results.

This enumeration can be expensive if we find many corrections of the individual terms, since we could encounter a large number of combinations of alternatives. Several heuristics are used to trim this space. In the example above, as we expand the alternatives for *flew* and *form*, we retain only the most frequent combinations in the collection or in the query logs, which contain previous queries by users. For instance, we would retain *flew* from as an alternative to try and extend to a three-term corrected query, but perhaps not *fled* fore or *flea* form. In this example, the biword *fled fore* is likely to be rare compared to the biword *flew from*. Then, we only attempt to extend the list of top biwords (such as *flew from*), to corrections of *Heathrow*. As an alternative to using the biword statistics in the collection, we may use the logs of queries issued by users; these could of course include queries with spelling errors.



Exercise 3.7

If $|s_i|$ denotes the length of string s_i , show that the edit distance between s_1 and s_2 is never more than $\max\{|s_1|, |s_2|\}$.

Exercise 3.8

Compute the edit distance between *paris* and *alice*. Write down the 5×5 array of distances between all prefixes as computed by the algorithm in Figure 3.5.

Exercise 3.9

Write pseudocode showing the details of computing on the fly the Jaccard coefficient while scanning the postings of the k -gram index, as mentioned on page 61.

Exercise 3.10

Compute the Jaccard coefficients between the query *bord* and each of the terms in Figure 3.7 that contain the bigram *or*.

Exercise 3.11

Consider the four-term query *atched in the rye* and suppose that each of the query terms has five alternative terms suggested by isolated-term correction. How many possible corrected phrases must we consider if we do not trim the space of corrected phrases, but instead try all six variants for each of the terms?

Exercise 3.12

For each of the prefixes of the query — *atched*, *atched in* and *atched in the* — we have a number of substitute prefixes arising from each term and its alternatives. Suppose that we were to retain only the top 10 of these substitute prefixes, as measured by its number of occurrences in the collection. We eliminate the rest from consideration for extension to longer prefixes: thus, if *atched in* is not one of the 10 most common 2-term queries in the collection, we do not consider any extension of *atched in* as possibly leading to a correction of *atched in the rye*. How many of the possible substitute prefixes are we eliminating at each phase?

Exercise 3.13

Are we guaranteed that retaining and extending only the 10 commonest substitute prefixes of *atched in* will lead to one of the 10 commonest substitute prefixes of *atched in the*?

3.4 Phonetic correction

Our final technique for tolerant retrieval has to do with *phonetic* correction: misspellings that arise because the user types a query that sounds like the target term. Such algorithms are especially applicable to searches on the names of people. The main idea here is to generate, for each term, a “phonetic hash” so that similar-sounding terms hash to the same value. The idea owes its origins to work in international police departments from the early 20th century, seeking to match names for wanted criminals despite the names being spelled differently in different countries. It is mainly used to correct phonetic misspellings in proper nouns.

SOUNDEX Algorithms for such phonetic hashing are commonly collectively known as *soundex* algorithms. However, there is an original soundex algorithm, with various variants, built on the following scheme:

1. Turn every term to be indexed into a 4-character reduced form. Build an inverted index from these reduced forms to the original terms; call this the soundex index.
2. Do the same with query terms.
3. When the query calls for a soundex match, search this soundex index.

The variations in different soundex algorithms have to do with the conversion of terms to 4-character forms. A commonly used conversion results in a 4-character code, with the first character being a letter of the alphabet and the other three being digits between 0 and 9.

1. Retain the first letter of the term.
2. Change all occurrences of the following letters to '0' (zero): 'A', 'E', 'I', 'O', 'U', 'H', 'W', 'Y'.
3. Change letters to digits as follows:
 - B, F, P, V to 1.
 - C, G, J, K, Q, S, X, Z to 2.
 - D, T to 3.
 - L to 4.
 - M, N to 5.
 - R to 6.
4. Repeatedly remove one out of each pair of consecutive identical digits.
5. Remove all zeros from the resulting string. Pad the resulting string with trailing zeros and return the first four positions, which will consist of a letter followed by three digits.

For an example of a soundex map, Hermann maps to H655. Given a query (say *herman*), we compute its soundex code and then retrieve all vocabulary terms matching this soundex code from the soundex index, before running the resulting query on the standard inverted index.

This algorithm rests on a few observations: (1) vowels are viewed as interchangeable, in transcribing names; (2) consonants with similar sounds (e.g., D and T) are put in equivalence classes. This leads to related names often having the same soundex codes. While these rules work for many cases, especially European languages, such rules tend to be writing system dependent. For example, Chinese names can be written in Wade-Giles or Pinyin transcription. While soundex works for some of the differences in the two transcriptions, for instance mapping both Wade-Giles hs and Pinyin x to 2, it fails in other cases, for example Wade-Giles j and Pinyin r are mapped differently.



Exercise 3.14

Find two differently spelled proper nouns whose soundex codes are the same.

Exercise 3.15

Find two phonetically similar proper nouns whose soundex codes are different.

3.5 References and further reading

[Knuth \(1997\)](#) is a comprehensive source for information on search trees, including B-trees and their use in searching through dictionaries.

[Garfield \(1976\)](#) gives one of the first complete descriptions of the permuterm index. [Ferragina and Venturini \(2007\)](#) give an approach to addressing the space blowup in permuterm indexes.

One of the earliest formal treatments of spelling correction was due to [Damerau \(1964\)](#). The notion of edit distance that we have used is due to [Levenshtein \(1965\)](#) and the algorithm in Figure 3.5 is due to [Wagner and Fischer \(1974\)](#). [Peterson \(1980\)](#) and [Kukich \(1992\)](#) developed variants of methods based on edit distances, culminating in a detailed empirical study of several methods by [Zobel and Dart \(1995\)](#), which shows that k -gram indexing is very effective for finding candidate mismatches, but should be combined with a more fine-grained technique such as edit distance to determine the most likely misspellings. [Gusfield \(1997\)](#) is a standard reference on string algorithms such as edit distance.

Probabilistic models (“noisy channel” models) for spelling correction were pioneered by [Kernighan et al. \(1990\)](#) and further developed by [Brill and Moore \(2000\)](#) and [Toutanova and Moore \(2002\)](#). In these models, the misspelled query is viewed as a probabilistic corruption of a correct query. They have a similar mathematical basis to the language model methods presented in Chapter 12, and also provide ways of incorporating phonetic similarity, closeness on the keyboard, and data from the actual spelling mistakes of users. Many would regard them as the state-of-the-art approach. [Cucerzan and Brill \(2004\)](#) show how this work can be extended to learning spelling correction models based on query reformulations in search engine logs.

The soundex algorithm is attributed to Margaret K. Odell and Robert C. Russelli (from U.S. patents granted in 1918 and 1922); the version described here draws on [Bourne and Ford \(1961\)](#). [Zobel and Dart \(1996\)](#) evaluate various phonetic matching algorithms, finding that a variant of the soundex algorithm performs poorly for general spelling correction, but that other algorithms based on the phonetic similarity of term pronunciations perform well.

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4 *Index construction*

INDEXING
INDEXER

In this chapter, we look at how to construct an inverted index. We call this process *index construction* or *indexing*; the process or machine that performs it the *indexer*. The design of indexing algorithms is governed by hardware constraints. We therefore begin this chapter with a review of the basics of computer hardware that are relevant for indexing. We then introduce blocked sort-based indexing (Section 4.2), an efficient single-machine algorithm designed for static collections that can be viewed as a more scalable version of the basic sort-based indexing algorithm we introduced in Chapter 1. Section 4.3 describes single-pass in-memory indexing, an algorithm that has even better scaling properties because it does not hold the vocabulary in memory. For very large collections like the web, indexing has to be distributed over computer clusters with hundreds or thousands of machines. We discuss this in Section 4.4. Collections with frequent changes require *dynamic indexing* introduced in Section 4.5 so that changes in the collection are immediately reflected in the index. Finally, we cover some complicating issues that can arise in indexing – such as security and indexes for ranked retrieval – in Section 4.6.

Index construction interacts with several topics covered in other chapters. The indexer needs raw text, but documents are encoded in many ways (see Chapter 2). Indexers compress and decompress intermediate files and the final index (see Chapter 5). In web search, documents are not on a local file system, but have to be spidered or crawled (see Chapter 20). In enterprise search, most documents are encapsulated in varied content management systems, email applications, and databases. We give some examples in Section 4.7. Although most of these applications can be accessed via http, native Application Programming Interfaces (APIs) are usually more efficient. The reader should be aware that building the subsystem that feeds raw text to the indexing process can in itself be a challenging problem.

► **Table 4.1** Typical system parameters in 2007. The seek time is the time needed to position the disk head in a new position. The transfer time per byte is the rate of transfer from disk to memory when the head is in the right position.

Symbol	Statistic	Value
s	average seek time	$5 \text{ ms} = 5 \times 10^{-3} \text{ s}$
b	transfer time per byte	$0.02 \mu\text{s} = 2 \times 10^{-8} \text{ s}$
	processor's clock rate	10^9 s^{-1}
p	lowlevel operation (e.g., compare & swap a word)	$0.01 \mu\text{s} = 10^{-8} \text{ s}$
	size of main memory	several GB
	size of disk space	1 TB or more

4.1 Hardware basics

When building an information retrieval (IR) system, many decisions are based on the characteristics of the computer hardware on which the system runs. We therefore begin this chapter with a brief review of computer hardware. Performance characteristics typical of systems in 2007 are shown in Table 4.1. A list of hardware basics that we need in this book to motivate IR system design follows.

CACHING

SEEK TIME

- Access to data in memory is much faster than access to data on disk. It takes a few clock cycles (perhaps 5×10^{-9} seconds) to access a byte in memory, but much longer to transfer it from disk (about 2×10^{-8} seconds). Consequently, we want to keep as much data as possible in memory, especially those data that we need to access frequently. We call the technique of keeping frequently used disk data in main memory *caching*.
- When doing a disk read or write, it takes a while for the disk head to move to the part of the disk where the data are located. This time is called the *seek time* and it averages 5 ms for typical disks. No data are being transferred during the seek. To maximize data transfer rates, chunks of data that will be read together should therefore be stored contiguously on disk. For example, using the numbers in Table 4.1 it may take as little as 0.2 seconds to transfer 10 megabytes (MB) from disk to memory if it is stored as one chunk, but up to $0.2 + 100 \times (5 \times 10^{-3}) = 0.7$ seconds if it is stored in 100 noncontiguous chunks because we need to move the disk head up to 100 times.
- Operating systems generally read and write entire blocks. Thus, reading a single byte from disk can take as much time as reading the entire block.

BUFFER

Block sizes of 8, 16, 32, and 64 kilobytes (KB) are common. We call the part of main memory where a block being read or written is stored a *buffer*.

- Data transfers from disk to memory are handled by the system bus, not by the processor. This means that the processor is available to process data during disk I/O. We can exploit this fact to speed up data transfers by storing compressed data on disk. Assuming an efficient decompression algorithm, the total time of reading and then decompressing compressed data is usually less than reading uncompressed data.
- Servers used in IR systems typically have several gigabytes (GB) of main memory, sometimes tens of GB. Available disk space is several orders of magnitude larger.

4.2 Blocked sort-based indexing

The basic steps in constructing a nonpositional index are depicted in Figure 1.4 (page 8). We first make a pass through the collection assembling all term–docID pairs. We then sort the pairs with the term as the dominant key and docID as the secondary key. Finally, we organize the docIDs for each term into a postings list and compute statistics like term and document frequency. For small collections, all this can be done in memory. In this chapter, we describe methods for large collections that require the use of secondary storage.

TERMID

To make index construction more efficient, we represent terms as termIDs (instead of strings as we did in Figure 1.4), where each *termID* is a unique serial number. We can build the mapping from terms to termIDs on the fly while we are processing the collection; or, in a two-pass approach, we compile the vocabulary in the first pass and construct the inverted index in the second pass. The index construction algorithms described in this chapter all do a single pass through the data. Section 4.7 gives references to multipass algorithms that are preferable in certain applications, for example, when disk space is scarce.

REUTERS-RCV1

We work with the *Reuters-RCV1* collection as our model collection in this chapter, a collection with roughly 1 GB of text. It consists of about 800,000 documents that were sent over the Reuters newswire during a 1-year period between August 20, 1996, and August 19, 1997. A typical document is shown in Figure 4.1, but note that we ignore multimedia information like images in this book and are only concerned with text. *Reuters-RCV1* covers a wide range of international topics, including politics, business, sports, and (as in this example) science. Some key statistics of the collection are shown in Table 4.2.

► **Table 4.2** Collection statistics for Reuters-RCV1. Values are rounded for the computations in this book. The unrounded values are: 806,791 documents, 222 tokens per document, 391,523 (distinct) terms, 6.04 bytes per token with spaces and punctuation, 4.5 bytes per token without spaces and punctuation, 7.5 bytes per term, and 96,969,056 tokens. The numbers in this table correspond to the third line (“case folding”) in Table 5.1 (page 87).

Symbol	Statistic	Value
N	documents	800,000
L_{ave}	avg. # tokens per document	200
M	terms	400,000
	avg. # bytes per token (incl. spaces/punct.)	6
	avg. # bytes per token (without spaces/punct.)	4.5
	avg. # bytes per term	7.5
T	tokens	100,000,000



► **Figure 4.1** Document from the Reuters newswire.

Reuters-RCV1 has 100 million tokens. Collecting all termID–docID pairs of the collection using 4 bytes each for termID and docID therefore requires 0.8 GB of storage. Typical collections today are often one or two orders of magnitude larger than Reuters-RCV1. You can easily see how such collections overwhelm even large computers if we try to sort their termID–docID pairs in memory. If the size of the intermediate files during index construction is within a small factor of available memory, then the compression techniques introduced in Chapter 5 can help; however, the postings file of many large collections cannot fit into memory even after compression.

With main memory insufficient, we need to use an *external sorting algorithm*, that is, one that uses disk. For acceptable speed, the central require-

```

BSBINDEXCONSTRUCTION()
1  $n \leftarrow 0$ 
2 while (all documents have not been processed)
3   do  $n \leftarrow n + 1$ 
4      $block \leftarrow \text{PARSENEXTBLOCK}()$ 
5     BSBI-INVERT( $block$ )
6     WRITEBLOCKTODISK( $block, f_n$ )
7   MERGEBLOCKS( $f_1, \dots, f_n; f_{\text{merged}}$ )

```

► **Figure 4.2** Blocked sort-based indexing. The algorithm stores inverted blocks in files f_1, \dots, f_n and the merged index in f_{merged} .

BLOCKED SORT-BASED INDEXING ALGORITHM

INVERSION

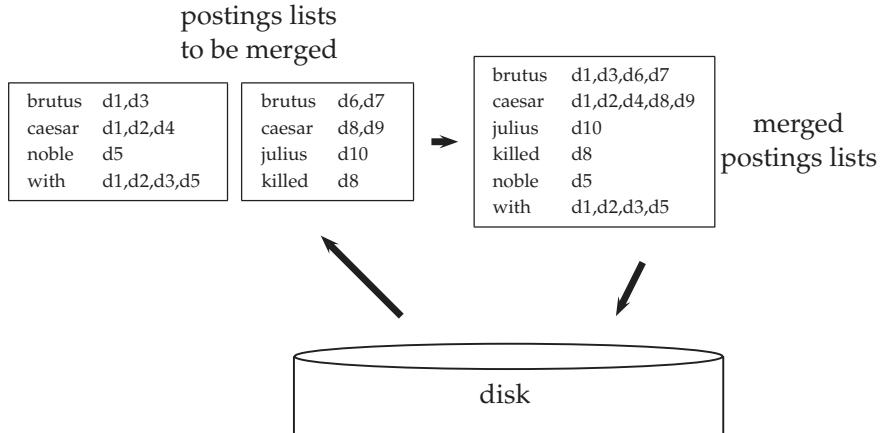
POSTING

ment of such an algorithm is that it minimize the number of random disk seeks during sorting – sequential disk reads are far faster than seeks as we explained in Section 4.1. One solution is the *blocked sort-based indexing algorithm* or *BSBI* in Figure 4.2. BSBI (i) segments the collection into parts of equal size, (ii) sorts the termID–docID pairs of each part in memory, (iii) stores intermediate sorted results on disk, and (iv) merges all intermediate results into the final index.

The algorithm parses documents into termID–docID pairs and accumulates the pairs in memory until a block of a fixed size is full (PARSENEXTBLOCK in Figure 4.2). We choose the block size to fit comfortably into memory to permit a fast in-memory sort. The block is then inverted and written to disk. *Inversion* involves two steps. First, we sort the termID–docID pairs. Next, we collect all termID–docID pairs with the same termID into a postings list, where a *posting* is simply a docID. The result, an inverted index for the block we have just read, is then written to disk. Applying this to Reuters-RCV1 and assuming we can fit 10 million termID–docID pairs into memory, we end up with ten blocks, each an inverted index of one part of the collection.

In the final step, the algorithm simultaneously merges the ten blocks into one large merged index. An example with two blocks is shown in Figure 4.3, where we use d_i to denote the i^{th} document of the collection. To do the merging, we open all block files simultaneously, and maintain small read buffers for the ten blocks we are reading and a write buffer for the final merged index we are writing. In each iteration, we select the lowest termID that has not been processed yet using a priority queue or a similar data structure. All postings lists for this termID are read and merged, and the merged list is written back to disk. Each read buffer is refilled from its file when necessary.

How expensive is BSBI? Its time complexity is $\Theta(T \log T)$ because the step with the highest time complexity is sorting and T is an upper bound for the number of items we must sort (i.e., the number of termID–docID pairs). But



► **Figure 4.3** Merging in blocked sort-based indexing. Two blocks (“postings lists to be merged”) are loaded from disk into memory, merged in memory (“merged postings lists”) and written back to disk. We show terms instead of termIDs for better readability.

the actual indexing time is usually dominated by the time it takes to parse the documents (PARSENEXTBLOCK) and to do the final merge (MERGEBLOCKS). Exercise 4.6 asks you to compute the total index construction time for RCV1 that includes these steps as well as inverting the blocks and writing them to disk.

Notice that Reuters-RCV1 is not particularly large in an age when one or more GB of memory are standard on personal computers. With appropriate compression (Chapter 5), we could have created an inverted index for RCV1 in memory on a not overly beefy server. The techniques we have described are needed, however, for collections that are several orders of magnitude larger.

?

Exercise 4.1

If we need $T \log_2 T$ comparisons (where T is the number of termID–docID pairs) and two disk seeks for each comparison, how much time would index construction for Reuters-RCV1 take if we used disk instead of memory for storage and an unoptimized sorting algorithm (i.e., not an external sorting algorithm)? Use the system parameters in Table 4.1.

Exercise 4.2

[*]

How would you create the dictionary in blocked sort-based indexing on the fly to avoid an extra pass through the data?

```

SPIMI-INVERT(token_stream)
1 output_file = NEWFILE()
2 dictionary = NEWHASH()
3 while (free memory available)
4   do token  $\leftarrow$  next(token_stream)
5     if term(token)  $\notin$  dictionary
6       then postings_list = ADDTODICTIONARY(dictionary, term(token))
7       else postings_list = GETPOSTINGSLIST(dictionary, term(token))
8     if full(postings_list)
9       then postings_list = DOUBLEPOSTINGSLIST(dictionary, term(token))
10      ADDTOPOSTINGSLIST(postings_list, docID(token))
11    sorted_terms  $\leftarrow$  SORTTERMS(dictionary)
12  WRITEBLOCKTODISK(sorted_terms, dictionary, output_file)
13  return output_file

```

► **Figure 4.4** Inversion of a block in single-pass in-memory indexing

4.3 Single-pass in-memory indexing

SINGLE-PASS
IN-MEMORY INDEXING

Blocked sort-based indexing has excellent scaling properties, but it needs a data structure for mapping terms to termIDs. For very large collections, this data structure does not fit into memory. A more scalable alternative is *single-pass in-memory indexing* or *SPIMI*. SPIMI uses terms instead of termIDs, writes each block's dictionary to disk, and then starts a new dictionary for the next block. SPIMI can index collections of any size as long as there is enough disk space available.

The SPIMI algorithm is shown in Figure 4.4. The part of the algorithm that parses documents and turns them into a stream of term–docID pairs, which we call *tokens* here, has been omitted. SPIMI-INVERT is called repeatedly on the token stream until the entire collection has been processed.

Tokens are processed one by one (line 4) during each successive call of SPIMI-INVERT. When a term occurs for the first time, it is added to the dictionary (best implemented as a hash), and a new postings list is created (line 6). The call in line 7 returns this postings list for subsequent occurrences of the term.

A difference between BSBI and SPIMI is that SPIMI adds a posting directly to its postings list (line 10). Instead of first collecting all termID–docID pairs and then sorting them (as we did in BSBI), each postings list is dynamic (i.e., its size is adjusted as it grows) and it is immediately available to collect postings. This has two advantages: It is faster because there is no sorting required, and it saves memory because we keep track of the term a postings

list belongs to, so the termIDs of postings need not be stored. As a result, the blocks that individual calls of SPIMI-INVERT can process are much larger and the index construction process as a whole is more efficient.

Because we do not know how large the postings list of a term will be when we first encounter it, we allocate space for a short postings list initially and double the space each time it is full (lines 8–9). This means that some memory is wasted, which counteracts the memory savings from the omission of termIDs in intermediate data structures. However, the overall memory requirements for the dynamically constructed index of a block in SPIMI are still lower than in BSBI.

When memory has been exhausted, we write the index of the block (which consists of the dictionary and the postings lists) to disk (line 12). We have to sort the terms (line 11) before doing this because we want to write postings lists in lexicographic order to facilitate the final merging step. If each block's postings lists were written in unsorted order, merging blocks could not be accomplished by a simple linear scan through each block.

Each call of SPIMI-INVERT writes a block to disk, just as in BSBI. The last step of SPIMI (corresponding to line 7 in Figure 4.2; not shown in Figure 4.4) is then to merge the blocks into the final inverted index.

In addition to constructing a new dictionary structure for each block and eliminating the expensive sorting step, SPIMI has a third important component: compression. Both the postings and the dictionary terms can be stored compactly on disk if we employ compression. Compression increases the efficiency of the algorithm further because we can process even larger blocks, and because the individual blocks require less space on disk. We refer readers to the literature for this aspect of the algorithm (Section 4.7).

The time complexity of SPIMI is $\Theta(T)$ because no sorting of tokens is required and all operations are at most linear in the size of the collection.

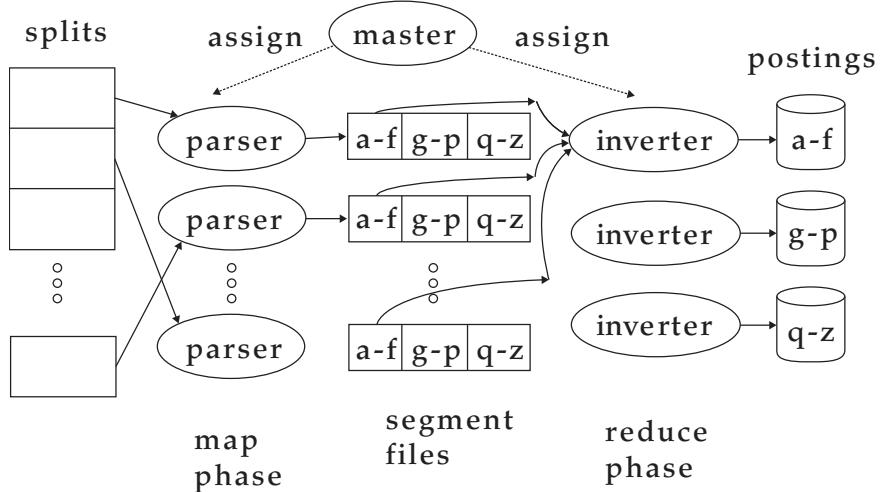
4.4 Distributed indexing

Collections are often so large that we cannot perform index construction efficiently on a single machine. This is particularly true of the World Wide Web for which we need large computer *clusters*¹ to construct any reasonably sized web index. Web search engines, therefore, use *distributed indexing* algorithms for index construction. The result of the construction process is a distributed index that is partitioned across several machines – either according to term or according to document. In this section, we describe distributed indexing for a term-partitioned index. Most large search engines prefer a document-

1. A cluster in this chapter is a group of tightly coupled computers that work together closely. This sense of the word is different from the use of cluster as a group of documents that are semantically similar in Chapters 16–18.

partitioned index (which can be easily generated from a term-partitioned index). We discuss this topic further in Section 20.3 (page 454).

MAPREDUCE	The distributed index construction method we describe in this section is an application of <i>MapReduce</i> , a general architecture for distributed computing. MapReduce is designed for large computer clusters. The point of a cluster is to solve large computing problems on cheap commodity machines or <i>nodes</i> that are built from standard parts (processor, memory, disk) as opposed to on a supercomputer with specialized hardware. Although hundreds or thousands of machines are available in such clusters, individual machines can fail at any time. One requirement for robust distributed indexing is, therefore, that we divide the work up into chunks that we can easily assign and – in case of failure – reassign. A <i>master node</i> directs the process of assigning and reassigning tasks to individual worker nodes.
MASTER NODE	
SPLITS	
KEY-VALUE PAIRS	The map and reduce phases of MapReduce split up the computing job into chunks that standard machines can process in a short time. The various steps of MapReduce are shown in Figure 4.5 and an example on a collection consisting of two documents is shown in Figure 4.6. First, the input data, in our case a collection of web pages, are split into <i>n splits</i> where the size of the split is chosen to ensure that the work can be distributed evenly (chunks should not be too large) and efficiently (the total number of chunks we need to manage should not be too large); 16 or 64 MB are good sizes in distributed indexing. Splits are not preassigned to machines, but are instead assigned by the master node on an ongoing basis: As a machine finishes processing one split, it is assigned the next one. If a machine dies or becomes a laggard due to hardware problems, the split it is working on is simply reassigned to another machine.
MAP PHASE	In general, MapReduce breaks a large computing problem into smaller parts by recasting it in terms of manipulation of <i>key-value pairs</i> . For indexing, a key-value pair has the form (termID,docID). In distributed indexing, the mapping from terms to termIDs is also distributed and therefore more complex than in single-machine indexing. A simple solution is to maintain a (perhaps precomputed) mapping for frequent terms that is copied to all nodes and to use terms directly (instead of termIDs) for infrequent terms. We do not address this problem here and assume that all nodes share a consistent term → termID mapping.
PARSER SEGMENT FILE	The <i>map phase</i> of MapReduce consists of mapping splits of the input data to key-value pairs. This is the same parsing task we also encountered in BSBI and SPIMI, and we therefore call the machines that execute the map phase <i>parsers</i> . Each parser writes its output to local intermediate files, the <i>segment files</i> (shown as a-f g-p q-z in Figure 4.5).
REDUCE PHASE	For the <i>reduce phase</i> , we want all values for a given key to be stored close together, so that they can be read and processed quickly. This is achieved by



► **Figure 4.5** An example of distributed indexing with MapReduce. Adapted from Dean and Ghemawat (2004).

partitioning the keys into j term partitions and having the parsers write key-value pairs for each term partition into a separate segment file. In Figure 4.5, the term partitions are according to first letter: a–f, g–p, q–z, and $j = 3$. (We chose these key ranges for ease of exposition. In general, key ranges need not correspond to contiguous terms or termIDs.) The term partitions are defined by the person who operates the indexing system (Exercise 4.10). The parsers then write corresponding segment files, one for each term partition. Each term partition thus corresponds to r segments files, where r is the number of parsers. For instance, Figure 4.5 shows three a–f segment files of the a–f partition, corresponding to the three parsers shown in the figure.

Collecting all values (here: docIDs) for a given key (here: termID) into one list is the task of the *inverters* in the reduce phase. The master assigns each term partition to a different inverter – and, as in the case of parsers, reassigns term partitions in case of failing or slow inverters. Each term partition (corresponding to r segment files, one on each parser) is processed by one inverter. We assume here that segment files are of a size that a single machine can handle (Exercise 4.9). Finally, the list of values is sorted for each key and written to the final sorted postings list (“postings” in the figure). (Note that postings in Figure 4.6 include term frequencies, whereas each posting in the other sections of this chapter is simply a docID without term frequency information.) The data flow is shown for a–f in Figure 4.5. This completes the construction of the inverted index.

Schema of map and reduce functions

map: input	\rightarrow list(k, v)
reduce: ($k, \text{list}(v)$)	\rightarrow output

Instantiation of the schema for index construction

map: web collection	\rightarrow list(termID, docID)
reduce: ($\langle \text{termID}_1, \text{list}(\text{docID}) \rangle, \langle \text{termID}_2, \text{list}(\text{docID}) \rangle, \dots$)	\rightarrow (postings_list ₁ , postings_list ₂ , ...)

Example for index construction

map: $d_2 : C$ died. $d_1 : C$ came, C c'ed.	\rightarrow ($\langle C, d_2 \rangle, \langle \text{died}, d_2 \rangle, \langle C, d_1 \rangle, \langle \text{came}, d_1 \rangle, \langle C, d_1 \rangle, \langle \text{c'ed}, d_1 \rangle$)
reduce: ($\langle C, (d_2, d_1, d_1) \rangle, \langle \text{died}, (d_2) \rangle, \langle \text{came}, (d_1) \rangle, \langle \text{c'ed}, (d_1) \rangle$)	\rightarrow ($\langle C, (d_1, 2, d_2, 1) \rangle, \langle \text{died}, (d_2, 1) \rangle, \langle \text{came}, (d_1, 1) \rangle, \langle \text{c'ed}, (d_1, 1) \rangle$)

► **Figure 4.6** Map and reduce functions in MapReduce. In general, the map function produces a list of key-value pairs. All values for a key are collected into one list in the reduce phase. This list is then processed further. The instantiations of the two functions and an example are shown for index construction. Because the map phase processes documents in a distributed fashion, termID–docID pairs need not be ordered correctly initially as in this example. The example shows terms instead of termIDs for better readability. We abbreviate Caesar as C and conquered as c'ed.

Parsers and inverters are not separate sets of machines. The master identifies idle machines and assigns tasks to them. The same machine can be a parser in the map phase and an inverter in the reduce phase. And there are often other jobs that run in parallel with index construction, so in between being a parser and an inverter a machine might do some crawling or another unrelated task.

To minimize write times before inverters reduce the data, each parser writes its segment files to its *local disk*. In the reduce phase, the master communicates to an inverter the locations of the relevant segment files (e.g., of the r segment files of the a-f partition). Each segment file only requires one sequential read because all data relevant to a particular inverter were written to a single segment file by the parser. This setup minimizes the amount of network traffic needed during indexing.

Figure 4.6 shows the general schema of the MapReduce functions. Input and output are often lists of key-value pairs themselves, so that several MapReduce jobs can run in sequence. In fact, this was the design of the Google indexing system in 2004. What we describe in this section corresponds to only one of five to ten MapReduce operations in that indexing system. Another MapReduce operation transforms the term-partitioned index we just created into a document-partitioned one.

MapReduce offers a robust and conceptually simple framework for implementing index construction in a distributed environment. By providing a semiautomatic method for splitting index construction into smaller tasks, it can scale to almost arbitrarily large collections, given computer clusters of

sufficient size.



Exercise 4.3

For $n = 15$ splits, $r = 10$ segments, and $j = 3$ term partitions, how long would distributed index creation take for Reuters-RCV1 in a MapReduce architecture? Base your assumptions about cluster machines on Table 4.1.

4.5 Dynamic indexing

Thus far, we have assumed that the document collection is static. This is fine for collections that change infrequently or never (e.g., the Bible or Shakespeare). But most collections are modified frequently with documents being added, deleted, and updated. This means that new terms need to be added to the dictionary, and postings lists need to be updated for existing terms.

The simplest way to achieve this is to periodically reconstruct the index from scratch. This is a good solution if the number of changes over time is small and a delay in making new documents searchable is acceptable – and if enough resources are available to construct a new index while the old one is still available for querying.

AUXILIARY INDEX

If there is a requirement that new documents be included quickly, one solution is to maintain two indexes: a large main index and a small *auxiliary index* that stores new documents. The auxiliary index is kept in memory. Searches are run across both indexes and results merged. Deletions are stored in an invalidation bit vector. We can then filter out deleted documents before returning the search result. Documents are updated by deleting and reinserting them.

Each time the auxiliary index becomes too large, we merge it into the main index. The cost of this merging operation depends on how we store the index in the file system. If we store each postings list as a separate file, then the merge simply consists of extending each postings list of the main index by the corresponding postings list of the auxiliary index. In this scheme, the reason for keeping the auxiliary index is to reduce the number of disk seeks required over time. Updating each document separately requires up to M_{ave} disk seeks, where M_{ave} is the average size of the vocabulary of documents in the collection. With an auxiliary index, we only put additional load on the disk when we merge auxiliary and main indexes.

Unfortunately, the one-file-per-postings-list scheme is infeasible because most file systems cannot efficiently handle very large numbers of files. The simplest alternative is to store the index as one large file, that is, as a concatenation of all postings lists. In reality, we often choose a compromise between the two extremes (Section 4.7). To simplify the discussion, we choose the simple option of storing the index as one large file here.

```

LMERGEADDTOKEN(indexes,  $Z_0$ , token)
1    $Z_0 \leftarrow \text{MERGE}(Z_0, \{\text{token}\})$ 
2   if  $|Z_0| = n$ 
3     then for  $i \leftarrow 0$  to  $\infty$ 
4       do if  $I_i \in \text{indexes}$ 
5         then  $Z_{i+1} \leftarrow \text{MERGE}(I_i, Z_i)$ 
6           ( $Z_{i+1}$  is a temporary index on disk.)
7            $\text{indexes} \leftarrow \text{indexes} - \{I_i\}$ 
8         else  $I_i \leftarrow Z_i$  ( $Z_i$  becomes the permanent index  $I_i$ .)
9            $\text{indexes} \leftarrow \text{indexes} \cup \{I_i\}$ 
10          BREAK
11    $Z_0 \leftarrow \emptyset$ 

LOGARITHMICMERGE()
1    $Z_0 \leftarrow \emptyset$  ( $Z_0$  is the in-memory index.)
2    $\text{indexes} \leftarrow \emptyset$ 
3   while true
4   do LMERGEADDTOKEN(indexes,  $Z_0$ , GETNEXTTOKEN())

```

► **Figure 4.7** Logarithmic merging. Each token (termID,docID) is initially added to in-memory index Z_0 by LMERGEADDTOKEN. LOGARITHMICMERGE initializes Z_0 and *indexes*.

In this scheme, we process each posting $\lfloor T/n \rfloor$ times because we touch it during each of $\lfloor T/n \rfloor$ merges where n is the size of the auxiliary index and T the total number of postings. Thus, the overall time complexity is $\Theta(T^2/n)$. (We neglect the representation of terms here and consider only the docIDs. For the purpose of time complexity, a postings list is simply a list of docIDs.)

LOGARITHMIC MERGING

We can do better than $\Theta(T^2/n)$ by introducing $\log_2(T/n)$ indexes I_0, I_1, I_2, \dots of size $2^0 \times n, 2^1 \times n, 2^2 \times n, \dots$. Postings percolate up this sequence of indexes and are processed only once on each level. This scheme is called *logarithmic merging* (Figure 4.7). As before, up to n postings are accumulated in an in-memory auxiliary index, which we call Z_0 . When the limit n is reached, the $2^0 \times n$ postings in Z_0 are transferred to a new index I_0 that is created on disk. The next time Z_0 is full, it is merged with I_0 to create an index Z_1 of size $2^1 \times n$. Then Z_1 is either stored as I_1 (if there isn't already an I_1) or merged with I_1 into Z_2 (if I_1 exists); and so on. We service search requests by querying in-memory Z_0 and all currently valid indexes I_i on disk and merging the results. Readers familiar with the binomial heap data structure² will recog-

2. See, for example, (Cormen et al. 1990, Chapter 19).

nize its similarity with the structure of the inverted indexes in logarithmic merging.

Overall index construction time is $\Theta(T \log(T/n))$ because each posting is processed only once on each of the $\log(T/n)$ levels. We trade this efficiency gain for a slow down of query processing; we now need to merge results from $\log(T/n)$ indexes as opposed to just two (the main and auxiliary indexes). As in the auxiliary index scheme, we still need to merge very large indexes occasionally (which slows down the search system during the merge), but this happens less frequently and the indexes involved in a merge on average are smaller.

Having multiple indexes complicates the maintenance of collection-wide statistics. For example, it affects the spelling correction algorithm in Section 3.3 (page 56) that selects the corrected alternative with the most hits. With multiple indexes and an invalidation bit vector, the correct number of hits for a term is no longer a simple lookup. In fact, all aspects of an IR system – index maintenance, query processing, distribution, and so on – are more complex in logarithmic merging.

Because of this complexity of dynamic indexing, some large search engines adopt a reconstruction-from-scratch strategy. They do not construct indexes dynamically. Instead, a new index is built from scratch periodically. Query processing is then switched from the new index and the old index is deleted.



Exercise 4.4

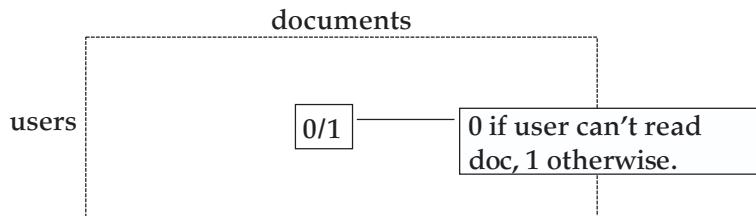
For $n = 2$ and $1 \leq T \leq 30$, perform a step-by-step simulation of the algorithm in Figure 4.7. Create a table that shows, for each point in time at which $T = 2 * k$ tokens have been processed ($1 \leq k \leq 15$), which of the three indexes I_0, \dots, I_3 are in use. The first three lines of the table are given below.

	I_3	I_2	I_1	I_0
2	0	0	0	0
4	0	0	0	1
6	0	0	1	0

4.6 Other types of indexes

This chapter only describes construction of nonpositional indexes. Except for the much larger data volume we need to accommodate, the main difference for positional indexes is that $(\text{termID}, \text{docID}, (\text{position}1, \text{position}2, \dots))$ triples, instead of $(\text{termID}, \text{docID})$ pairs have to be processed and that tokens and postings contain positional information in addition to docIDs. With this change, the algorithms discussed here can all be applied to positional indexes.

In the indexes we have considered so far, postings lists are ordered with respect to docID. As we see in Chapter 5, this is advantageous for compres-



► **Figure 4.8** A user-document matrix for access control lists. Element (i, j) is 1 if user i has access to document j and 0 otherwise. During query processing, a user's access postings list is intersected with the results list returned by the text part of the index.

RANKED RETRIEVAL SYSTEMS	
SECURITY	
ACCESS CONTROL LISTS	

sion – instead of docIDs we can compress smaller *gaps* between IDs, thus reducing space requirements for the index. However, this structure for the index is not optimal when we build *ranked* (Chapters 6 and 7) – as opposed to Boolean – *retrieval systems*. In ranked retrieval, postings are often ordered according to weight or impact, with the highest-weighted postings occurring first. With this organization, scanning of long postings lists during query processing can usually be terminated early when weights have become so small that any further documents can be predicted to be of low similarity to the query (see Chapter 6). In a docID-sorted index, new documents are always inserted at the end of postings lists. In an impact-sorted index (Section 7.1.5, page 140), the insertion can occur anywhere, thus complicating the update of the inverted index.

Security is an important consideration for retrieval systems in corporations. A low-level employee should not be able to find the salary roster of the corporation, but authorized managers need to be able to search for it. Users' results lists must not contain documents they are barred from opening; the very existence of a document can be sensitive information.

User authorization is often mediated through *access control lists* or ACLs. ACLs can be dealt with in an information retrieval system by representing each document as the set of users that can access them (Figure 4.8) and then inverting the resulting user-document matrix. The inverted ACL index has, for each user, a “postings list” of documents they can access – the user’s access list. Search results are then intersected with this list. However, such an index is difficult to maintain when access permissions change – we discussed these difficulties in the context of incremental indexing for regular postings lists in Section 4.5. It also requires the processing of very long postings lists for users with access to large document subsets. User membership is therefore often verified by retrieving access information directly from the file system at query time – even though this slows down retrieval.

► **Table 4.3** The five steps in constructing an index for Reuters-RCV1 in blocked sort-based indexing. Line numbers refer to Figure 4.2.

Step	Time
1 reading of collection (line 4)	
2 10 initial sorts of 10^7 records each (line 5)	
3 writing of 10 blocks (line 6)	
4 total disk transfer time for merging (line 7)	
5 time of actual merging (line 7)	
total	

► **Table 4.4** Collection statistics for a large collection.

Symbol	Statistic	Value
N	# documents	1,000,000,000
L_{ave}	# tokens per document	1000
M	# distinct terms	44,000,000

We discussed indexes for storing and retrieving terms (as opposed to documents) in Chapter 3.



Exercise 4.5

Can spelling correction compromise document-level security? Consider the case where a spelling correction is based on documents to which the user does not have access.



Exercise 4.6

Total index construction time in blocked sort-based indexing is broken down in Table 4.3. Fill out the time column of the table for Reuters-RCV1 assuming a system with the parameters given in Table 4.1.

Exercise 4.7

Repeat Exercise 4.6 for the larger collection in Table 4.4. Choose a block size that is realistic for current technology (remember that a block should easily fit into main memory). How many blocks do you need?

Exercise 4.8

Assume that we have a collection of modest size whose index can be constructed with the simple in-memory indexing algorithm in Figure 1.4 (page 8). For this collection, compare memory, disk and time requirements of the simple algorithm in Figure 1.4 and blocked sort-based indexing.

Exercise 4.9

Assume that machines in MapReduce have 100 GB of disk space each. Assume further that the postings list of the term has a size of 200 GB. Then the MapReduce algorithm as described cannot be run to construct the index. How would you modify MapReduce so that it can handle this case?

Exercise 4.10

For optimal load balancing, the inverters in MapReduce must get segmented postings files of similar sizes. For a new collection, the distribution of key-value pairs may not be known in advance. How would you solve this problem?

Exercise 4.11

Apply MapReduce to the problem of counting how often each term occurs in a set of files. Specify map and reduce operations for this task. Write down an example along the lines of Figure 4.6.

Exercise 4.12

We claimed (on page 80) that an auxiliary index can impair the quality of collection statistics. An example is the term weighting method idf, which is defined as $\log(N/df_i)$ where N is the total number of documents and df_i is the number of documents that term i occurs in (Section 6.2.1, page 117). Show that even a small auxiliary index can cause significant error in idf when it is computed on the main index only. Consider a rare term that suddenly occurs frequently (e.g., Flossie as in Tropical Storm Flossie).

4.7 References and further reading

Witten et al. (1999, Chapter 5) present an extensive treatment of the subject of index construction and additional indexing algorithms with different trade-offs of memory, disk space, and time. In general, blocked sort-based indexing does well on all three counts. However, if conserving memory or disk space is the main criterion, then other algorithms may be a better choice. See Witten et al. (1999), Tables 5.4 and 5.5; BSBI is closest to “sort-based multiway merge,” but the two algorithms differ in dictionary structure and use of compression.

Moffat and Bell (1995) show how to construct an index “in situ,” that is, with disk space usage close to what is needed for the final index and with a minimum of additional temporary files (cf. also Harman and Candela (1990)). They give Lesk (1988) and Somogyi (1990) credit for being among the first to employ sorting for index construction.

The SPIMI method in Section 4.3 is from (Heinz and Zobel 2003). We have simplified several aspects of the algorithm, including compression and the fact that each term’s data structure also contains, in addition to the postings list, its document frequency and house keeping information. We recommend Heinz and Zobel (2003) and Zobel and Moffat (2006) as up-to-date, in-depth treatments of index construction. Other algorithms with good scaling properties with respect to vocabulary size require several passes through the data, e.g., FAST-INV (Fox and Lee 1991, Harman et al. 1992).

The MapReduce architecture was introduced by Dean and Ghemawat (2004).

An open source implementation of MapReduce is available at <http://lucene.apache.org/hadoop/>. Ribeiro-Neto et al. (1999) and Melnik et al. (2001) describe other approaches

to distributed indexing. Introductory chapters on distributed IR are (Baeza-Yates and Ribeiro-Neto 1999, Chapter 9) and (Grossman and Frieder 2004, Chapter 8). See also Callan (2000).

Lester et al. (2005) and Büttcher and Clarke (2005a) analyze the properties of logarithmic merging and compare it with other construction methods. One of the first uses of this method was in Lucene (<http://lucene.apache.org>). Other dynamic indexing methods are discussed by Büttcher et al. (2006) and Lester et al. (2006). The latter paper also discusses the strategy of replacing the old index by one built from scratch.

Heinz et al. (2002) compare data structures for accumulating the vocabulary in memory. Büttcher and Clarke (2005b) discuss security models for a common inverted index for multiple users. A detailed characterization of the Reuters-RCV1 collection can be found in (Lewis et al. 2004). NIST distributes the collection (see <http://trec.nist.gov/data/reuters/reuters.html>).

Garcia-Molina et al. (1999, Chapter 2) review computer hardware relevant to system design in depth.

An effective indexer for enterprise search needs to be able to communicate efficiently with a number of applications that hold text data in corporations, including Microsoft Outlook, IBM's Lotus software, databases like Oracle and MySQL, content management systems like Open Text, and enterprise resource planning software like SAP.

5

Index compression

Chapter 1 introduced the dictionary and the inverted index as the central data structures in information retrieval (IR). In this chapter, we employ a number of compression techniques for dictionary and inverted index that are essential for efficient IR systems.

One benefit of compression is immediately clear. We need less disk space. As we will see, compression ratios of 1:4 are easy to achieve, potentially cutting the cost of storing the index by 75%.

There are two more subtle benefits of compression. The first is increased use of caching. Search systems use some parts of the dictionary and the index much more than others. For example, if we cache the postings list of a frequently used query term t , then the computations necessary for responding to the one-term query t can be entirely done in memory. With compression, we can fit a lot more information into main memory. Instead of having to expend a disk seek when processing a query with t , we instead access its postings list in memory and decompress it. As we will see below, there are simple and efficient decompression methods, so that the penalty of having to decompress the postings list is small. As a result, we are able to decrease the response time of the IR system substantially. Because memory is a more expensive resource than disk space, increased speed owing to caching – rather than decreased space requirements – is often the prime motivator for compression.

The second more subtle advantage of compression is faster transfer of data from disk to memory. Efficient decompression algorithms run so fast on modern hardware that the total time of transferring a compressed chunk of data from disk and then decompressing it is usually less than transferring the same chunk of data in uncompressed form. For instance, we can reduce input/output (I/O) time by loading a much smaller compressed postings list, even when you add on the cost of decompression. So, in most cases, the retrieval system runs faster on compressed postings lists than on uncompressed postings lists.

If the main goal of compression is to conserve disk space, then the speed

of compression algorithms is of no concern. But for improved cache utilization and faster disk-to-memory transfer, decompression speeds must be high. The compression algorithms we discuss in this chapter are highly efficient and can therefore serve all three purposes of index compression.

POSTING

In this chapter, we define a *posting* as a docID in a postings list. For example, the postings list (6; 20, 45, 100), where 6 is the termID of the list's term, contains three postings. As discussed in Section 2.4.2 (page 41), postings in most search systems also contain frequency and position information; but we will only consider simple docID postings here. See Section 5.4 for references on compressing frequencies and positions.

This chapter first gives a statistical characterization of the distribution of the entities we want to compress – terms and postings in large collections (Section 5.1). We then look at compression of the dictionary, using the dictionary-as-a-string method and blocked storage (Section 5.2). Section 5.3 describes two techniques for compressing the postings file, variable byte encoding and γ encoding.

5.1 Statistical properties of terms in information retrieval

As in the last chapter, we use Reuters-RCV1 as our model collection (see Table 4.2, page 70). We give some term and postings statistics for the collection in Table 5.1. “ $\Delta\%$ ” indicates the reduction in size from the previous line. “T%” is the cumulative reduction from unfiltered.

The table shows the number of terms for different levels of preprocessing (column 2). The number of terms is the main factor in determining the size of the dictionary. The number of nonpositional postings (column 3) is an indicator of the expected size of the nonpositional index of the collection. The expected size of a positional index is related to the number of positions it must encode (column 4).

RULE OF 30

In general, the statistics in Table 5.1 show that preprocessing affects the size of the dictionary and the number of nonpositional postings greatly. Stemming and case folding reduce the number of (distinct) terms by 17% each and the number of nonpositional postings by 4% and 3%, respectively. The treatment of the most frequent words is also important. The *rule of 30* states that the 30 most common words account for 30% of the tokens in written text (31% in the table). Eliminating the 150 most common words from indexing (as stop words; cf. Section 2.2.2, page 27) cuts 25% to 30% of the nonpositional postings. But, although a stop list of 150 words reduces the number of postings by a quarter or more, this size reduction does not carry over to the size of the compressed index. As we will see later in this chapter, the postings lists of frequent words require only a few bits per posting after compression.

The deltas in the table are in a range typical of large collections. Note,

► **Table 5.1** The effect of preprocessing on the number of terms, nonpositional postings, and tokens for Reuters-RCV1. “Δ%” indicates the reduction in size from the previous line, except that “30 stop words” and “150 stop words” both use “case folding” as their reference line. “T%” is the cumulative (“total”) reduction from unfiltered. We performed stemming with the Porter stemmer (Chapter 2, page 33).

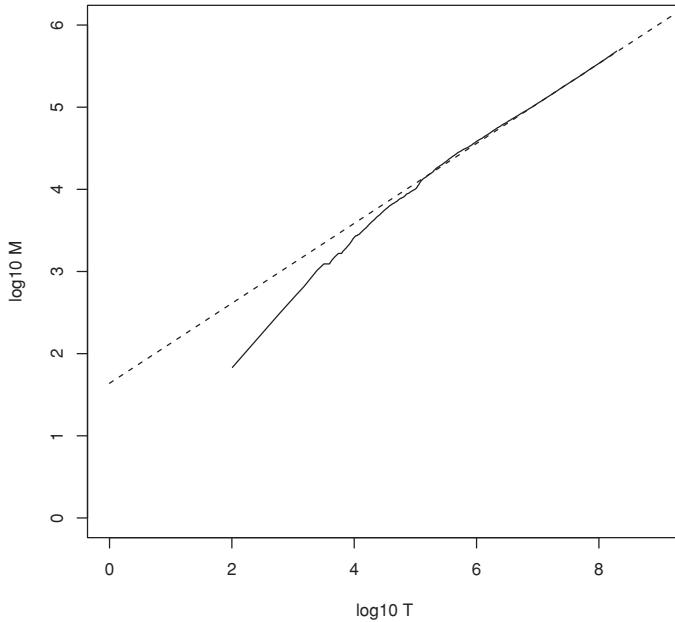
	(distinct) terms			nonpositional postings			tokens (= number of positive entries in postings)		
	number	Δ%	T%	number	Δ%	T%	number	Δ%	T%
unfiltered	484,494			109,971,179			197,879,290		
no numbers	473,723	−2	−2	100,680,242	−8	−8	179,158,204	−9	−9
case folding	391,523	−17	−19	96,969,056	−3	−12	179,158,204	−0	−9
30 stop words	391,493	−0	−19	83,390,443	−14	−24	121,857,825	−31	−38
150 stop words	391,373	−0	−19	67,001,847	−30	−39	94,516,599	−47	−52
stemming	322,383	−17	−33	63,812,300	−4	−42	94,516,599	−0	−52

however, that the percentage reductions can be very different for some text collections. For example, for a collection of web pages with a high proportion of French text, a lemmatizer for French reduces vocabulary size much more than the Porter stemmer does for an English-only collection because French is a morphologically richer language than English.

LOSSLESS LOSSY COMPRESSION

The compression techniques we describe in the remainder of this chapter are *lossless*, that is, all information is preserved. Better compression ratios can be achieved with *lossy compression*, which discards some information. Case folding, stemming, and stop word elimination are forms of lossy compression. Similarly, the vector space model (Chapter 6) and dimensionality reduction techniques like latent semantic indexing (Chapter 18) create compact representations from which we cannot fully restore the original collection. Lossy compression makes sense when the “lost” information is unlikely ever to be used by the search system. For example, web search is characterized by a large number of documents, short queries, and users who only look at the first few pages of results. As a consequence, we can discard postings of documents that would only be used for hits far down the list. Thus, there are retrieval scenarios where lossy methods can be used for compression without any reduction in effectiveness.

Before introducing techniques for compressing the dictionary, we want to estimate the number of distinct terms M in a collection. It is sometimes said that languages have a vocabulary of a certain size. The second edition of the *Oxford English Dictionary* (OED) defines more than 600,000 words. But the vocabulary of most large collections is much larger than the OED. The OED does not include most names of people, locations, products, or scientific



► **Figure 5.1** Heaps' law. Vocabulary size M as a function of collection size T (number of tokens) for Reuters-RCV1. For these data, the dashed line $\log_{10} M = 0.49 * \log_{10} T + 1.64$ is the best least-squares fit. Thus, $k = 10^{1.64} \approx 44$ and $b = 0.49$.

entities like genes. These names need to be included in the inverted index, so our users can search for them.

5.1.1 Heaps' law: Estimating the number of terms

HEAPS' LAW A better way of getting a handle on M is *Heaps' law*, which estimates vocabulary size as a function of collection size:

$$(5.1) \quad M = kT^b$$

where T is the number of tokens in the collection. Typical values for the parameters k and b are: $30 \leq k \leq 100$ and $b \approx 0.5$. The motivation for Heaps' law is that the simplest possible relationship between collection size and vocabulary size is linear in log–log space and the assumption of linearity is usually born out in practice as shown in Figure 5.1 for Reuters-RCV1. In this case, the fit is excellent for $T > 10^5 = 100,000$, for the parameter values $b = 0.49$ and $k = 44$. For example, for the first 1,000,020 tokens Heaps' law

predicts 38,323 terms:

$$44 \times 1,000,020^{0.49} \approx 38,323.$$

The actual number is 38,365 terms, very close to the prediction.

The parameter k is quite variable because vocabulary growth depends a lot on the nature of the collection and how it is processed. Case-folding and stemming reduce the growth rate of the vocabulary, whereas including numbers and spelling errors increase it. Regardless of the values of the parameters for a particular collection, Heaps' law suggests that (i) the dictionary size continues to increase with more documents in the collection, rather than a maximum vocabulary size being reached, and (ii) the size of the dictionary is quite large for large collections. These two hypotheses have been empirically shown to be true of large text collections (Section 5.4). So dictionary compression is important for an effective information retrieval system.

5.1.2 Zipf's law: Modeling the distribution of terms

We also want to understand how terms are distributed across documents. This helps us to characterize the properties of the algorithms for compressing postings lists in Section 5.3.

ZIPF'S LAW

A commonly used model of the distribution of terms in a collection is *Zipf's law*. It states that, if t_1 is the most common term in the collection, t_2 is the next most common, and so on, then the collection frequency cf_i of the i th most common term is proportional to $1/i$:

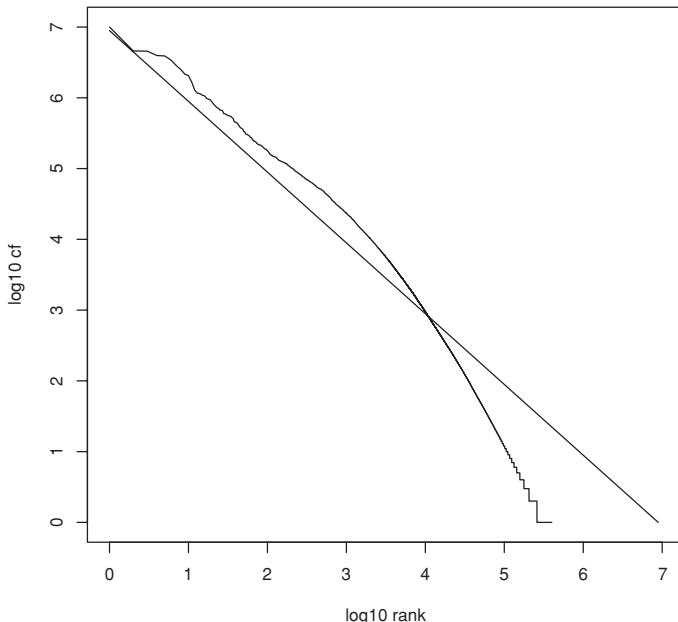
$$(5.2) \quad cf_i \propto \frac{1}{i}.$$

So if the most frequent term occurs cf_1 times, then the second most frequent term has half as many occurrences, the third most frequent term a third as many occurrences, and so on. The intuition is that frequency decreases very rapidly with rank. Equation (5.2) is one of the simplest ways of formalizing such a rapid decrease and it has been found to be a reasonably good model.

POWER LAW

Equivalently, we can write Zipf's law as $cf_i = ci^k$ or as $\log cf_i = \log c + k \log i$ where $k = -1$ and c is a constant to be defined in Section 5.3.2. It is therefore a *power law* with exponent $k = -1$. See Chapter 19, page 426, for another power law, a law characterizing the distribution of links on web pages.

The log–log graph in Figure 5.2 plots the collection frequency of a term as a function of its rank for Reuters-RCV1. A line with slope -1 , corresponding to the Zipf function $\log cf_i = \log c - \log i$, is also shown. The fit of the data to the law is not particularly good, but good enough to serve as a model for term distributions in our calculations in Section 5.3.



► **Figure 5.2** Zipf’s law for Reuters-RCV1. Frequency is plotted as a function of frequency rank for the terms in the collection. The line is the distribution predicted by Zipf’s law (weighted least-squares fit; intercept is 6.95).



Exercise 5.1

[*]

Assuming one machine word per posting, what is the size of the uncompressed (non-positional) index for different tokenizations based on Table 5.1? How do these numbers compare with Table 5.6?

5.2 Dictionary compression

This section presents a series of dictionary data structures that achieve increasingly higher compression ratios. The dictionary is small compared with the postings file as suggested by Table 5.1. So why compress it if it is responsible for only a small percentage of the overall space requirements of the IR system?

One of the primary factors in determining the response time of an IR system is the number of disk seeks necessary to process a query. If parts of the dictionary are on disk, then many more disk seeks are necessary in query evaluation. Thus, the main goal of compressing the dictionary is to fit it in main memory, or at least a large portion of it, to support high query through-

term	document frequency	pointer to postings list
a	656,265	→
aachen	65	→
...
zulu	221	→
space needed:		20 bytes 4 bytes 4 bytes

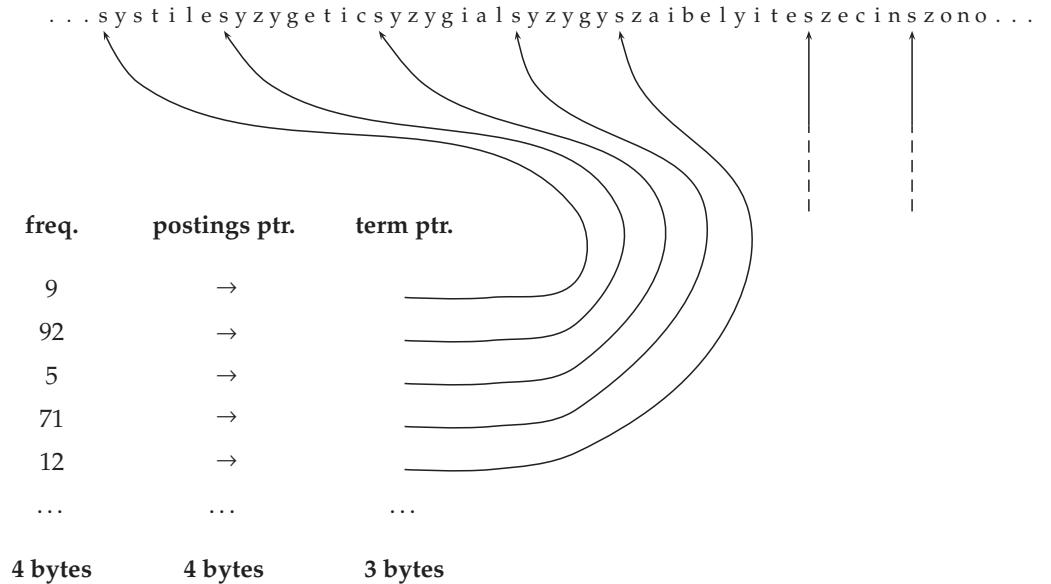
► **Figure 5.3** Storing the dictionary as an array of fixed-width entries.

put. Although dictionaries of very large collections fit into the memory of a standard desktop machine, this is not true of many other application scenarios. For example, an enterprise search server for a large corporation may have to index a multiterabyte collection with a comparatively large vocabulary because of the presence of documents in many different languages. We also want to be able to design search systems for limited hardware such as mobile phones and onboard computers. Other reasons for wanting to conserve memory are fast startup time and having to share resources with other applications. The search system on your PC must get along with the memory-hogging word processing suite you are using at the same time.

5.2.1 Dictionary as a string

The simplest data structure for the dictionary is to sort the vocabulary lexicographically and store it in an array of fixed-width entries as shown in Figure 5.3. We allocate 20 bytes for the term itself (because few terms have more than twenty characters in English), 4 bytes for its document frequency, and 4 bytes for the pointer to its postings list. Four-byte pointers resolve a 4 gigabytes (GB) address space. For large collections like the web, we need to allocate more bytes per pointer. We look up terms in the array by binary search. For Reuters-RCV1, we need $M \times (20 + 4 + 4) = 400,000 \times 28 = 11.2\text{megabytes (MB)}$ for storing the dictionary in this scheme.

Using fixed-width entries for terms is clearly wasteful. The average length of a term in English is about eight characters (Table 4.2, page 70), so on average we are wasting twelve characters in the fixed-width scheme. Also, we have no way of storing terms with more than twenty characters like hydrochlorofluorocarbons and supercalifragilisticexpialidocious. We can overcome these shortcomings by storing the dictionary terms as one long string of characters, as shown in Figure 5.4. The pointer to the next term is also used to demarcate the end of the current term. As before, we locate terms in the data structure by way of binary search in the (now smaller) table. This scheme saves us 60% compared to fixed-width storage – 12 bytes on average of the



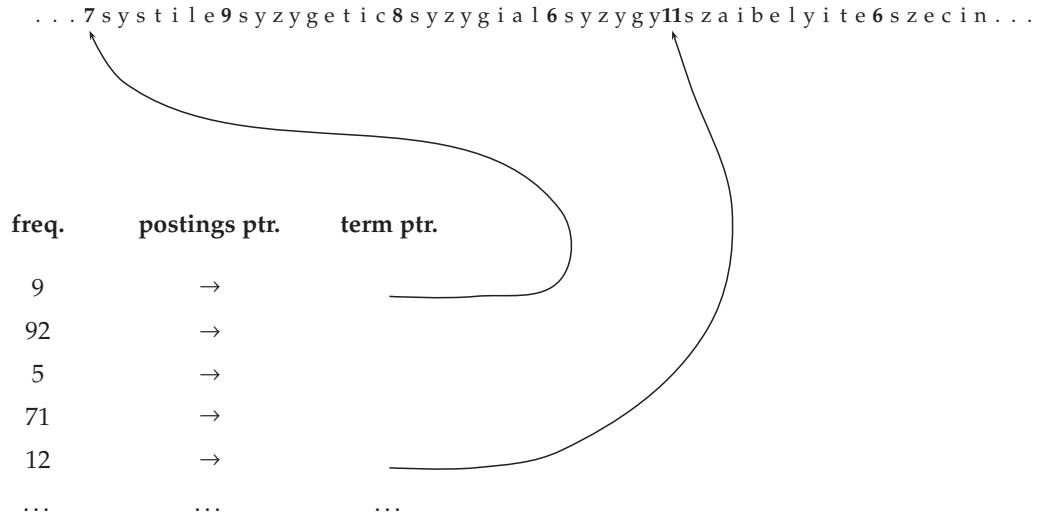
► **Figure 5.4** Dictionary-as-a-string storage. Pointers mark the end of the preceding term and the beginning of the next. For example, the first three terms in this example are `systile`, `syzygetic`, and `syzygial`.

20 bytes we allocated for terms before. However, we now also need to store term pointers. The term pointers resolve $400,000 \times 8 = 3.2 \times 10^6$ positions, so they need to be $\log_2 3.2 \times 10^6 \approx 22$ bits or 3 bytes long.

In this new scheme, we need $400,000 \times (4 + 4 + 3 + 8) = 7.6$ MB for the Reuters-RCV1 dictionary: 4 bytes each for frequency and postings pointer, 3 bytes for the term pointer, and 8 bytes on average for the term. So we have reduced the space requirements by one third from 11.2 to 7.6 MB.

5.2.2 Blocked storage

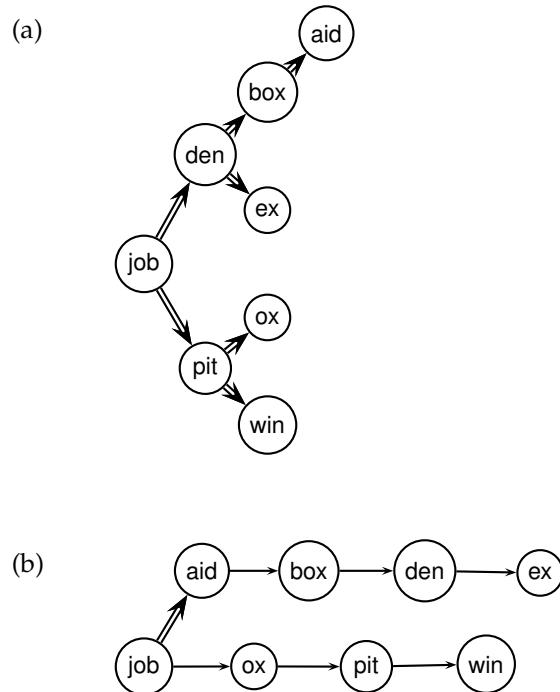
We can further compress the dictionary by grouping terms in the string into blocks of size k and keeping a term pointer only for the first term of each block (Figure 5.5). We store the length of the term in the string as an additional byte at the beginning of the term. We thus eliminate $k - 1$ term pointers, but need an additional k bytes for storing the length of each term. For $k = 4$, we save $(k - 1) \times 3 = 9$ bytes for term pointers, but need an additional $k = 4$ bytes for term lengths. So the total space requirements for the dictionary of Reuters-RCV1 are reduced by 5 bytes per four-term block, or a total of $400,000 \times 1/4 \times 5 = 0.5$ MB, bringing us down to 7.1 MB.



► **Figure 5.5** Blocked storage with four terms per block. The first block consists of *systile*, *syzygetic*, *syzygial*, and *syzygy* with lengths of seven, nine, eight, and six characters, respectively. Each term is preceded by a byte encoding its length that indicates how many bytes to skip to reach subsequent terms.

By increasing the block size k , we get better compression. However, there is a tradeoff between compression and the speed of term lookup. For the eight-term dictionary in Figure 5.6, steps in binary search are shown as double lines and steps in list search as simple lines. We search for terms in the uncompressed dictionary by binary search (a). In the compressed dictionary, we first locate the term’s block by binary search and then its position within the list by linear search through the block (b). Searching the uncompressed dictionary in (a) takes on average $(0 + 1 + 2 + 3 + 2 + 1 + 2 + 2)/8 \approx 1.6$ steps, assuming each term is equally likely to come up in a query. For example, finding the two terms, *aid* and *box*, takes three and two steps, respectively. With blocks of size $k = 4$ in (b), we need $(0 + 1 + 2 + 3 + 4 + 1 + 2 + 3)/8 = 2$ steps on average, $\approx 25\%$ more. For example, finding *den* takes one binary search step and two steps through the block. By increasing k , we can get the size of the compressed dictionary arbitrarily close to the minimum of $400,000 \times (4 + 4 + 1 + 8) = 6.8$ MB, but term lookup becomes prohibitively slow for large values of k .

One source of redundancy in the dictionary we have not exploited yet is the fact that consecutive entries in an alphabetically sorted list share common prefixes. This observation leads to *front coding* (Figure 5.7). A common prefix



► **Figure 5.6** Search of the uncompressed dictionary (a) and a dictionary compressed by blocking with $k = 4$ (b).

One block in blocked compression ($k = 4$) ...
 8automata8automate9automatic10automation

↓

... further compressed with front coding.
 8automat*a1◊e2◊ic3◊ion

► **Figure 5.7** Front coding. A sequence of terms with identical prefix (“automat”) is encoded by marking the end of the prefix with * and replacing it with ◊ in subsequent terms. As before, the first byte of each entry encodes the number of characters.

► **Table 5.2** Dictionary compression for Reuters-RCV1.

data structure	size in MB
dictionary, fixed-width	11.2
dictionary, term pointers into string	7.6
~, with blocking, $k = 4$	7.1
~, with blocking & front coding	5.9

is identified for a subsequence of the term list and then referred to with a special character. In the case of Reuters, front coding saves another 1.2 MB, as we found in an experiment.

Other schemes with even greater compression rely on minimal perfect hashing, that is, a hash function that maps M terms onto $[1, \dots, M]$ without collisions. However, we cannot adapt perfect hashes incrementally because each new term causes a collision and therefore requires the creation of a new perfect hash function. Therefore, they cannot be used in a dynamic environment.

Even with the best compression scheme, it may not be feasible to store the entire dictionary in main memory for very large text collections and for hardware with limited memory. If we have to partition the dictionary onto pages that are stored on disk, then we can index the first term of each page using a B-tree. For processing most queries, the search system has to go to disk anyway to fetch the postings. One additional seek for retrieving the term's dictionary page from disk is a significant, but tolerable increase in the time it takes to process a query.

Table 5.2 summarizes the compression achieved by the four dictionary data structures.



Exercise 5.2

Estimate the space usage of the Reuters-RCV1 dictionary with blocks of size $k = 8$ and $k = 16$ in blocked dictionary storage.

Exercise 5.3

Estimate the time needed for term lookup in the compressed dictionary of Reuters-RCV1 with block sizes of $k = 4$ (Figure 5.6, b), $k = 8$, and $k = 16$. What is the slowdown compared with $k = 1$ (Figure 5.6, a)?

5.3 Postings file compression

Recall from Table 4.2 (page 70) that Reuters-RCV1 has 800,000 documents, 200 tokens per document, six characters per token, and 100,000,000 postings where we define a posting in this chapter as a docID in a postings list, that is, excluding frequency and position information. These numbers

► **Table 5.3** Encoding gaps instead of document IDs. For example, we store gaps 107, 5, 43, ..., instead of docIDs 283154, 283159, 283202, ... for computer. The first docID is left unchanged (only shown for arachnocentric).

	encoding	postings list				
the	docIDs	...	283042	283043	283044	283045
	gaps			1	1	1
computer	docIDs	...	283047	283154	283159	283202
	gaps			107	5	43
arachnocentric	docIDs	252000	500100			
	gaps	252000	248100			

correspond to line 3 (“case folding”) in Table 5.1. Document identifiers are $\log_2 800,000 \approx 20$ bits long. Thus, the size of the collection is about $800,000 \times 200 \times 6$ bytes = 960 MB and the size of the uncompressed postings file is $100,000,000 \times 20/8 = 250$ MB.

To devise a more efficient representation of the postings file, one that uses fewer than 20 bits per document, we observe that the postings for frequent terms are close together. Imagine going through the documents of a collection one by one and looking for a frequent term like computer. We will find a document containing computer, then we skip a few documents that do not contain it, then there is again a document with the term and so on (see Table 5.3). The key idea is that the *gaps* between postings are short, requiring a lot less space than 20 bits to store. In fact, gaps for the most frequent terms such as the and for are mostly equal to 1. But the gaps for a rare term that occurs only once or twice in a collection (e.g., arachnocentric in Table 5.3) have the same order of magnitude as the docIDs and need 20 bits. For an economical representation of this distribution of gaps, we need a *variable encoding* method that uses fewer bits for short gaps.

To encode small numbers in less space than large numbers, we look at two types of methods: bytewise compression and bitwise compression. As the names suggest, these methods attempt to encode gaps with the minimum number of bytes and bits, respectively.

5.3.1 Variable byte codes

VARIABLE BYTE
ENCODING
CONTINUATION BIT

Variable byte (VB) encoding uses an integral number of bytes to encode a gap. The last 7 bits of a byte are “payload” and encode part of the gap. The first bit of the byte is a *continuation bit*. It is set to 1 for the last byte of the encoded gap and to 0 otherwise. To decode a variable byte code, we read a sequence of bytes with continuation bit 0 terminated by a byte with continuation bit 1. We then extract and concatenate the 7-bit parts. Figure 5.8 gives pseudocode

```

VBENCODENUMBER( $n$ )
1  $bytes \leftarrow \langle \rangle$ 
2 while  $true$ 
3 do PREPEND( $bytes, n \bmod 128$ )
4   if  $n < 128$ 
5     then BREAK
6    $n \leftarrow n \text{ div } 128$ 
7  $bytes[\text{LENGTH}(bytes)] += 128$ 
8 return  $bytes$ 

VBENCODE( $numbers$ )
1  $bytestream \leftarrow \langle \rangle$ 
2 for each  $n \in numbers$ 
3 do  $bytes \leftarrow \text{VBENCODENUMBER}(n)$ 
4    $bytestream \leftarrow \text{EXTEND}(bytestream, bytes)$ 
5 return  $bytestream$ 

VBDECODE( $bytestream$ )
1  $numbers \leftarrow \langle \rangle$ 
2  $n \leftarrow 0$ 
3 for  $i \leftarrow 1$  to  $\text{LENGTH}(bytestream)$ 
4 do if  $bytestream[i] < 128$ 
5   then  $n \leftarrow 128 \times n + bytestream[i]$ 
6   else  $n \leftarrow 128 \times n + (bytestream[i] - 128)$ 
7   APPEND( $numbers, n$ )
8    $n \leftarrow 0$ 
9 return  $numbers$ 

```

► **Figure 5.8** VB encoding and decoding. The functions div and mod compute integer division and remainder after integer division, respectively. PREPEND adds an element to the beginning of a list, for example, PREPEND($\langle 1, 2 \rangle, 3$) = $\langle 3, 1, 2 \rangle$. EXTEND extends a list, for example, EXTEND($\langle 1, 2 \rangle, \langle 3, 4 \rangle$) = $\langle 1, 2, 3, 4 \rangle$.

► **Table 5.4** VB encoding. Gaps are encoded using an integral number of bytes. The first bit, the continuation bit, of each byte indicates whether the code ends with this byte (1) or not (0).

docIDs	824	829	215406
gaps		5	214577
VB code	00000110 10111000	10000101	00001101 00001100 10110001

► **Table 5.5** Some examples of unary and γ codes. Unary codes are only shown for the smaller numbers. Commas in γ codes are for readability only and are not part of the actual codes.

number	unary code	length	offset	γ code
0	0			
1	10	0		0
2	110	10	0	10,0
3	1110	10	1	10,1
4	11110	110	00	110,00
9	1111111110	1110	001	1110,001
13		1110	101	1110,101
24		11110	1000	11110,1000
511		111111110	11111111	111111110,11111111
1025		1111111110	0000000001	1111111110,0000000001

for VB encoding and decoding and Table 5.4 an example of a VB-encoded postings list.¹

With VB compression, the size of the compressed index for Reuters-RCV1 is 116 MB as we verified in an experiment. This is a more than 50% reduction of the size of the uncompressed index (see Table 5.6).

The idea of VB encoding can also be applied to larger or smaller units than bytes: 32-bit words, 16-bit words, and 4-bit words or *nibbles*. Larger words further decrease the amount of bit manipulation necessary at the cost of less effective (or no) compression. Word sizes smaller than bytes get even better compression ratios at the cost of more bit manipulation. In general, bytes offer a good compromise between compression ratio and speed of decompression.

For most IR systems variable byte codes offer an excellent tradeoff between time and space. They are also simple to implement – most of the alternatives referred to in Section 5.4 are more complex. But if disk space is a scarce resource, we can achieve better compression ratios by using bit-level encodings, in particular two closely related encodings: γ codes, which we will turn to next, and δ codes (Exercise 5.9).



5.3.2 γ codes

VB codes use an adaptive number of *bytes* depending on the size of the gap. Bit-level codes adapt the length of the code on the finer grained *bit* level. The

1. Note that the origin is 0 in the table. Because we never need to encode a docID or a gap of 0, in practice the origin is usually 1, so that 10000000 encodes 1, 10000101 encodes 6 (not 5 as in the table), and so on.

UNARY CODE

simplest bit-level code is *unary code*. The unary code of n is a string of n 1s followed by a 0 (see the first two columns of Table 5.5). Obviously, this is not a very efficient code, but it will come in handy in a moment.

How efficient can a code be in principle? Assuming the 2^n gaps G with $1 \leq G \leq 2^n$ are all equally likely, the optimal encoding uses n bits for each G . So some gaps ($G = 2^n$ in this case) cannot be encoded with fewer than $\log_2 G$ bits. Our goal is to get as close to this lower bound as possible.

 γ ENCODING

A method that is within a factor of optimal is γ *encoding*. γ codes implement variable-length encoding by splitting the representation of a gap G into a pair of *length* and *offset*. *Offset* is G in binary, but with the leading 1 removed.² For example, for 13 (binary 1101) *offset* is 101. *Length* encodes the length of *offset* in unary code. For 13, the length of *offset* is 3 bits, which is 1110 in unary. The γ code of 13 is therefore 1110101, the concatenation of length 1110 and offset 101. The right hand column of Table 5.5 gives additional examples of γ codes.

A γ code is decoded by first reading the unary code up to the 0 that terminates it, for example, the four bits 1110 when decoding 1110101. Now we know how long the offset is: 3 bits. The offset 101 can then be read correctly and the 1 that was chopped off in encoding is prepended: 101 → 1101 = 13.

The length of *offset* is $\lfloor \log_2 G \rfloor$ bits and the length of *length* is $\lfloor \log_2 G \rfloor + 1$ bits, so the length of the entire code is $2 \times \lfloor \log_2 G \rfloor + 1$ bits. γ codes are always of odd length and they are within a factor of 2 of what we claimed to be the optimal encoding length $\log_2 G$. We derived this optimum from the assumption that the 2^n gaps between 1 and 2^n are equiprobable. But this need not be the case. In general, we do not know the probability distribution over gaps a priori.

ENTROPY

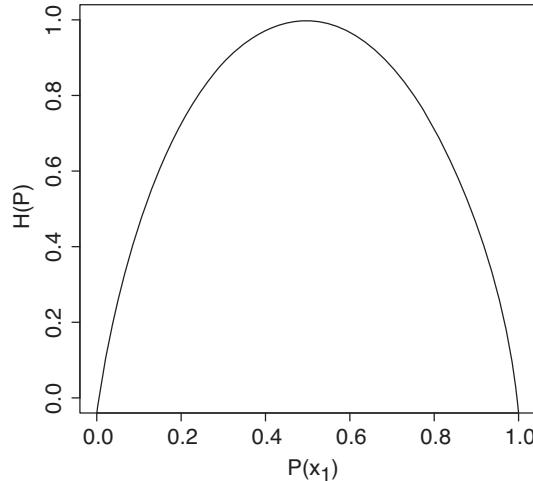
The characteristic of a discrete probability distribution³ P that determines its coding properties (including whether a code is optimal) is its *entropy* $H(P)$, which is defined as follows:

$$H(P) = - \sum_{x \in X} P(x) \log_2 P(x)$$

where X is the set of all possible numbers we need to be able to encode (and therefore $\sum_{x \in X} P(x) = 1.0$). Entropy is a measure of uncertainty as shown in Figure 5.9 for a probability distribution P over two possible outcomes, namely, $X = \{x_1, x_2\}$. Entropy is maximized ($H(P) = 1$) for $P(x_1) = P(x_2) = 0.5$ when uncertainty about which x_i will appear next is largest; and

2. We assume here that G has no leading 0s. If there are any, they are removed before deleting the leading 1.

3. Readers who want to review basic concepts of probability theory may want to consult Rice (2006) or Ross (2006). Note that we are interested in probability distributions over integers (gaps, frequencies, etc.), but that the coding properties of a probability distribution are independent of whether the outcomes are integers or something else.



► **Figure 5.9** Entropy $H(P)$ as a function of $P(x_1)$ for a sample space with two outcomes x_1 and x_2 .

minimized ($H(P) = 0$) for $P(x_1) = 1, P(x_2) = 0$ and for $P(x_1) = 0, P(x_2) = 1$ when there is absolute certainty.

It can be shown that the lower bound for the expected length $E(L)$ of a code L is $H(P)$ if certain conditions hold (see the references). It can further be shown that for $1 < H(P) < \infty$, γ encoding is within a factor of 3 of this optimal encoding, approaching 2 for large $H(P)$:

$$\frac{E(L_\gamma)}{H(P)} \leq 2 + \frac{1}{H(P)} \leq 3.$$

What is remarkable about this result is that it holds for any probability distribution P . So without knowing anything about the properties of the distribution of gaps, we can apply γ codes and be certain that they are within a factor of ≈ 2 of the optimal code for distributions of large entropy. A code like γ code with the property of being within a factor of optimal for an arbitrary distribution P is called *universal*.

UNIVERSAL CODE

PREFIX FREE

PARAMETER FREE

In addition to universality, γ codes have two other properties that are useful for index compression. First, they are *prefix free*, namely, no γ code is the prefix of another. This means that there is always a unique decoding of a sequence of γ codes – and we do not need delimiters between them, which would decrease the efficiency of the code. The second property is that γ codes are *parameter free*. For many other efficient codes, we have to fit the parameters of a model (e.g., the binomial distribution) to the distribution

of gaps in the index. This complicates the implementation of compression and decompression. For instance, the parameters need to be stored and retrieved. And in dynamic indexing, the distribution of gaps can change, so that the original parameters are no longer appropriate. These problems are avoided with a parameter-free code.

How much compression of the inverted index do γ codes achieve? To answer this question we use Zipf's law, the term distribution model introduced in Section 5.1.2. According to Zipf's law, the collection frequency cf_i is proportional to the inverse of the rank i , that is, there is a constant c' such that:

$$(5.3) \quad cf_i = \frac{c'}{i}.$$

We can choose a different constant c such that the fractions c/i are relative frequencies and sum to 1 (that is, $c/i = cf_i/T$):

$$(5.4) \quad 1 = \sum_{i=1}^M \frac{c}{i} = c \sum_{i=1}^M \frac{1}{i} = c H_M$$

$$(5.5) \quad c = \frac{1}{H_M}$$

where M is the number of distinct terms and H_M is the M th harmonic number.⁴ Reuters-RCV1 has $M = 400,000$ distinct terms and $H_M \approx \ln M$, so we have

$$c = \frac{1}{H_M} \approx \frac{1}{\ln M} = \frac{1}{\ln 400,000} \approx \frac{1}{13}.$$

Thus the i th term has a relative frequency of roughly $1/(13i)$, and the expected average number of occurrences of term i in a document of length L is:

$$L \frac{c}{i} \approx \frac{200 \times \frac{1}{13}}{i} \approx \frac{15}{i}$$

where we interpret the relative frequency as a term occurrence probability. Recall that 200 is the average number of tokens per document in Reuters-RCV1 (Table 4.2).

Now we have derived term statistics that characterize the distribution of terms in the collection and, by extension, the distribution of gaps in the postings lists. From these statistics, we can calculate the space requirements for an inverted index compressed with γ encoding. We first stratify the vocabulary into blocks of size $L_c = 15$. On average, term i occurs $15/i$ times per

4. Note that, unfortunately, the conventional symbol for both entropy and harmonic number is H . Context should make clear which is meant in this chapter.

	N documents
Lc most frequent terms	N gaps of 1 each
Lc next most frequent terms	$N/2$ gaps of 2 each
Lc next most frequent terms	$N/3$ gaps of 3 each
...	...

► **Figure 5.10** Stratification of terms for estimating the size of a γ encoded inverted index.

document. So the average number of occurrences \bar{f} per document is $1 \leq \bar{f}$ for terms in the first block, corresponding to a total number of N gaps per term. The average is $\frac{1}{2} \leq \bar{f} < 1$ for terms in the second block, corresponding to $N/2$ gaps per term, and $\frac{1}{3} \leq \bar{f} < \frac{1}{2}$ for terms in the third block, corresponding to $N/3$ gaps per term, and so on. (We take the lower bound because it simplifies subsequent calculations. As we will see, the final estimate is too pessimistic, even with this assumption.) We will make the somewhat unrealistic assumption that all gaps for a given term have the same size as shown in Figure 5.10. Assuming such a uniform distribution of gaps, we then have gaps of size 1 in block 1, gaps of size 2 in block 2, and so on.

Encoding the N/j gaps of size j with γ codes, the number of bits needed for the postings list of a term in the j th block (corresponding to one row in the figure) is:

$$\begin{aligned} \text{bits-per-row} &= \frac{N}{j} \times (2 \times \lfloor \log_2 j \rfloor + 1) \\ &\approx \frac{2N \log_2 j}{j}. \end{aligned}$$

To encode the entire block, we need $(Lc) \cdot (2N \log_2 j)/j$ bits. There are $M/(Lc)$ blocks, so the postings file as a whole will take up:

$$(5.6) \quad \sum_{j=1}^{\frac{M}{Lc}} \frac{2NLc \log_2 j}{j}.$$

► **Table 5.6** Index and dictionary compression for Reuters-RCV1. The compression ratio depends on the proportion of actual text in the collection. Reuters-RCV1 contains a large amount of XML markup. Using the two best compression schemes, γ encoding and blocking with front coding, the ratio compressed index to collection size is therefore especially small for Reuters-RCV1: $(101 + 5.9)/3600 \approx 0.03$.

data structure	size in MB
dictionary, fixed-width	11.2
dictionary, term pointers into string	7.6
\sim , with blocking, $k = 4$	7.1
\sim , with blocking & front coding	5.9
collection (text, xml markup etc)	3600.0
collection (text)	960.0
term incidence matrix	40,000.0
postings, uncompressed (32-bit words)	400.0
postings, uncompressed (20 bits)	250.0
postings, variable byte encoded	116.0
postings, γ encoded	101.0

For Reuters-RCV1, $\frac{M}{L_c} \approx 400,000/15 \approx 27,000$ and

$$(5.7) \quad \sum_{j=1}^{27,000} \frac{2 \times 10^6 \times 15 \log_2 j}{j} \approx 224 \text{ MB.}$$

So the postings file of the compressed inverted index for our 960 MB collection has a size of 224 MB, one fourth the size of the original collection.

When we run γ compression on Reuters-RCV1, the actual size of the compressed index is even lower: 101 MB, a bit more than one tenth of the size of the collection. The reason for the discrepancy between predicted and actual value is that (i) Zipf's law is not a very good approximation of the actual distribution of term frequencies for Reuters-RCV1 and (ii) gaps are not uniform. The Zipf model predicts an index size of 251 MB for the unrounded numbers from Table 4.2. If term frequencies are generated from the Zipf model and a compressed index is created for these artificial terms, then the compressed size is 254 MB. So to the extent that the assumptions about the distribution of term frequencies are accurate, the predictions of the model are correct.

Table 5.6 summarizes the compression techniques covered in this chapter. The term incidence matrix (Figure 1.1, page 4) for Reuters-RCV1 has size $400,000 \times 800,000 = 40 \times 8 \times 10^9$ bits or 40 GB.

γ codes achieve great compression ratios – about 15% better than variable byte codes for Reuters-RCV1. But they are expensive to decode. This is because many bit-level operations – shifts and masks – are necessary to decode a sequence of γ codes as the boundaries between codes will usually be

somewhere in the middle of a machine word. As a result, query processing is more expensive for γ codes than for variable byte codes. Whether we choose variable byte or γ encoding depends on the characteristics of an application, for example, on the relative weights we give to conserving disk space versus maximizing query response time.

The compression ratio for the index in Table 5.6 is about 25%: 400 MB (uncompressed, each posting stored as a 32-bit word) versus 101 MB (γ) and 116 MB (VB). This shows that both γ and VB codes meet the objectives we stated in the beginning of the chapter. Index compression substantially improves time and space efficiency of indexes by reducing the amount of disk space needed, increasing the amount of information that can be kept in the cache, and speeding up data transfers from disk to memory.



Exercise 5.4

[\star]

Compute variable byte codes for the numbers in Tables 5.3 and 5.5.

Exercise 5.5

[\star]

Compute variable byte and γ codes for the postings list $\langle 777, 17743, 294068, 31251336 \rangle$. Use gaps instead of docIDs where possible. Write binary codes in 8-bit blocks.

Exercise 5.6

Consider the postings list $\langle 4, 10, 11, 12, 15, 62, 63, 265, 268, 270, 400 \rangle$ with a corresponding list of gaps $\langle 4, 6, 1, 1, 3, 47, 1, 202, 3, 2, 130 \rangle$. Assume that the length of the postings list is stored separately, so the system knows when a postings list is complete. Using variable byte encoding: (i) What is the largest gap you can encode in 1 byte? (ii) What is the largest gap you can encode in 2 bytes? (iii) How many bytes will the above postings list require under this encoding? (Count only space for encoding the sequence of numbers.)

Exercise 5.7

A little trick is to notice that a gap cannot be of length 0 and that the stuff left to encode after shifting cannot be 0. Based on these observations: (i) Suggest a modification to variable byte encoding that allows you to encode slightly larger gaps in the same amount of space. (ii) What is the largest gap you can encode in 1 byte? (iii) What is the largest gap you can encode in 2 bytes? (iv) How many bytes will the postings list in Exercise 5.6 require under this encoding? (Count only space for encoding the sequence of numbers.)

Exercise 5.8

[\star]

From the following sequence of γ -coded gaps, reconstruct first the gap sequence and then the postings sequence: 11100011010101111101101111011.

Exercise 5.9

 δ CODES

γ codes are relatively inefficient for large numbers (e.g., 1025 in Table 5.5) as they encode the length of the offset in inefficient unary code. δ codes differ from γ codes in that they encode the first part of the code (*length*) in γ code instead of unary code. The encoding of *offset* is the same. For example, the δ code of 7 is 10,0,11 (again, we add commas for readability). 10,0 is the γ code for *length* (2 in this case) and the encoding of *offset* (11) is unchanged. (i) Compute the δ codes for the other numbers

► **Table 5.7** Two gap sequences to be merged in blocked sort-based indexing

γ encoded gap sequence of run 1 11101101111100101111111110100011111001

γ encoded gap sequence of run 2 1111101000011111100010001111110010000011111010101

in Table 5.5. For what range of numbers is the δ code shorter than the γ code? (ii) γ code beats variable byte code in Table 5.6 because the index contains stop words and thus many small gaps. Show that variable byte code is more compact if larger gaps dominate. (iii) Compare the compression ratios of δ code and variable byte code for a distribution of gaps dominated by large gaps.

Exercise 5.10

Go through the above calculation of index size and explicitly state all the approximations that were made to arrive at Equation (5.6).

Exercise 5.11

For a collection of your choosing, determine the number of documents and terms and the average length of a document. (i) How large is the inverted index predicted to be by Equation (5.6)? (ii) Implement an indexer that creates a γ -compressed inverted index for the collection. How large is the actual index? (iii) Implement an indexer that uses variable byte encoding. How large is the variable byte encoded index?

Exercise 5.12

To be able to hold as many postings as possible in main memory, it is a good idea to compress intermediate index files during index construction. (i) This makes merging runs in blocked sort-based indexing more complicated. As an example, work out the γ -encoded merged sequence of the gaps in Table 5.7. (ii) Index construction is more space efficient when using compression. Would you also expect it to be faster?

Exercise 5.13

(i) Show that the size of the vocabulary is finite according to Zipf's law and infinite according to Heaps' law. (ii) Can we derive Heaps' law from Zipf's law?

5.4 References and further reading

Heaps' law was discovered by [Heaps \(1978\)](#). See also [Baeza-Yates and Ribeiro-Neto \(1999\)](#). A detailed study of vocabulary growth in large collections is [\(Williams and Zobel 2005\)](#). Zipf's law is due to [Zipf \(1949\)](#). [Witten and Bell \(1990\)](#) investigate the quality of the fit obtained by the law. Other term distribution models, including K mixture and two-poisson model, are discussed by [Manning and Schütze \(1999, Chapter 15\)](#). [Carmel et al. \(2001\)](#), [Büttcher and Clarke \(2006\)](#), [Blanco and Barreiro \(2007\)](#), and [Ntoulas and Cho \(2007\)](#) show that lossy compression can achieve good compression with no or no significant decrease in retrieval effectiveness.

Dictionary compression is covered in detail by [Witten et al. \(1999, Chapter 4\)](#), which is recommended as additional reading.

Subsection 5.3.1 is based on (Scholer et al. 2002). The authors find that variable byte codes process queries two times faster than either bit-level compressed indexes or uncompressed indexes with a 30% penalty in compression ratio compared with the best bit-level compression method. They also show that compressed indexes can be superior to uncompressed indexes not only in disk usage, but also in query processing speed. Compared with VB codes, “variable nibble” codes showed 5% to 10% better compression and up to one third worse effectiveness in one experiment (Anh and Moffat 2005). Trotman (2003) also recommends using VB codes unless disk space is at a premium. In recent work, Anh and Moffat (2005; 2006a) and Zukowski et al. (2006) have constructed word-aligned binary codes that are both faster in decompression and at least as efficient as VB codes. Zhang et al. (2007) investigate the increased effectiveness of caching when a number of different compression techniques for postings lists are used on modern hardware.

δ codes (Exercise 5.9) and γ codes were introduced by Elias (1975), who proved that both codes are universal. In addition, δ codes are asymptotically optimal for $H(P) \rightarrow \infty$. δ codes perform better than γ codes if large numbers (greater than 15) dominate. A good introduction to information theory, including the concept of entropy, is (Cover and Thomas 1991). While Elias codes are only asymptotically optimal, arithmetic codes (Witten et al. 1999, Section 2.4) can be constructed to be arbitrarily close to the optimum $H(P)$ for any P .

Several additional index compression techniques are covered by Witten et al. (1999; Sections 3.3 and 3.4 and Chapter 5). They recommend using *parameterized codes* for index compression, codes that explicitly model the probability distribution of gaps for each term. For example, they show that *Golomb codes* achieve better compression ratios than γ codes for large collections. Moffat and Zobel (1992) compare several parameterized methods, including LLRUN (Fraenkel and Klein 1985).

The distribution of gaps in a postings list depends on the assignment of docIDs to documents. A number of researchers have looked into assigning docIDs in a way that is conducive to the efficient compression of gap sequences (Moffat and Stuiver 1996; Blandford and Bleloch 2002; Silvestri et al. 2004; Blanco and Barreiro 2006; Silvestri 2007). These techniques assign docIDs in a small range to documents in a cluster where a cluster can consist of all documents in a given time period, on a particular web site, or sharing another property. As a result, when a sequence of documents from a cluster occurs in a postings list, their gaps are small and can be more effectively compressed.

Different considerations apply to the compression of term frequencies and word positions than to the compression of docIDs in postings lists. See Scholer et al. (2002) and Zobel and Moffat (2006). Zobel and Moffat (2006) is recommended in general as an in-depth and up-to-date tutorial on inverted

indexes, including index compression.

This chapter only looks at index compression for Boolean retrieval. For ranked retrieval (Chapter 6), it is advantageous to order postings according to term frequency instead of docID. During query processing, the scanning of many postings lists can then be terminated early because smaller weights do not change the ranking of the highest ranked k documents found so far. It is not a good idea to precompute and store weights in the index (as opposed to frequencies) because they cannot be compressed as well as integers (see Section 7.1.5, page 140).

Document compression can also be important in an efficient information retrieval system. [de Moura et al. \(2000\)](#) and [Brisaboa et al. \(2007\)](#) describe compression schemes that allow direct searching of terms and phrases in the compressed text, which is infeasible with standard text compression utilities like gzip and compress.



Exercise 5.14

[\star]

We have defined unary codes as being “10”: sequences of 1s terminated by a 0. Interchanging the roles of 0s and 1s yields an equivalent “01” unary code. When this 01 unary code is used, the construction of a γ code can be stated as follows: (1) Write G down in binary using $b = \lfloor \log_2 j \rfloor + 1$ bits. (2) Prepend $(b - 1)$ 0s. (i) Encode the numbers in Table 5.5 in this alternative γ code. (ii) Show that this method produces a well-defined alternative γ code in the sense that it has the same length and can be uniquely decoded.

Exercise 5.15

[$\star\star\star$]

Unary code is not a universal code in the sense defined above. However, there exists a distribution over gaps for which unary code is optimal. Which distribution is this?

Exercise 5.16

Give some examples of terms that violate the assumption that gaps all have the same size (which we made when estimating the space requirements of a γ -encoded index). What are general characteristics of these terms?

Exercise 5.17

Consider a term whose postings list has size n , say, $n = 10,000$. Compare the size of the γ -compressed gap-encoded postings list if the distribution of the term is uniform (i.e., all gaps have the same size) versus its size when the distribution is not uniform. Which compressed postings list is smaller?

Exercise 5.18

Work out the sum in Equation (5.7) and show it adds up to about 251 MB. Use the numbers in Table 4.2, but do not round L_c , c , and the number of vocabulary blocks.

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6 *Scoring, term weighting and the vector space model*

Thus far we have dealt with indexes that support Boolean queries: a document either matches or does not match a query. In the case of large document collections, the resulting number of matching documents can far exceed the number a human user could possibly sift through. Accordingly, it is essential for a search engine to rank-order the documents matching a query. To do this, the search engine computes, for each matching document, a score with respect to the query at hand. In this chapter we initiate the study of assigning a score to a (query, document) pair. This chapter consists of three main ideas.

1. We introduce parametric and zone indexes in Section 6.1, which serve two purposes. First, they allow us to index and retrieve documents by metadata such as the language in which a document is written. Second, they give us a simple means for scoring (and thereby ranking) documents in response to a query.
2. Next, in Section 6.2 we develop the idea of weighting the importance of a term in a document, based on the statistics of occurrence of the term.
3. In Section 6.3 we show that by viewing each document as a vector of such weights, we can compute a score between a query and each document. This view is known as vector space scoring.

Section 6.4 develops several variants of term-weighting for the vector space model. Chapter 7 develops computational aspects of vector space scoring, and related topics.

As we develop these ideas, the notion of a query will assume multiple nuances. In Section 6.1 we consider queries in which specific query terms occur in specified regions of a matching document. Beginning Section 6.2 we will in fact relax the requirement of matching specific regions of a document; instead, we will look at so-called free text queries that simply consist of query terms with no specification on their relative order, importance or where in a document they should be found. The bulk of our study of scoring will be in this latter notion of a query being such a set of terms.

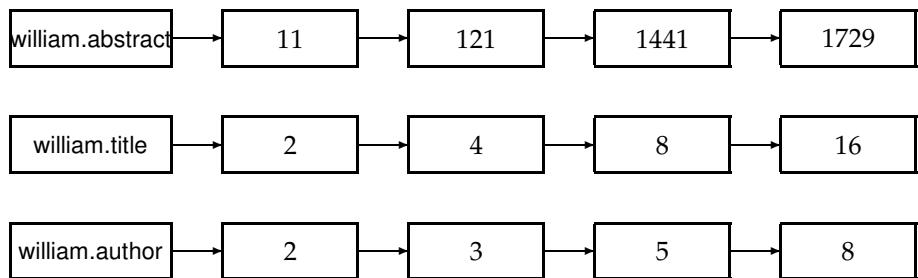
6.1 Parametric and zone indexes

METADATA	We have thus far viewed a document as a sequence of terms. In fact, most documents have additional structure. Digital documents generally encode, in machine-recognizable form, certain <i>metadata</i> associated with each document. By metadata, we mean specific forms of data about a document, such as its author(s), title and date of publication. This metadata would generally include <i>fields</i> such as the date of creation and the format of the document, as well the author and possibly the title of the document. The possible values of a field should be thought of as finite – for instance, the set of all dates of authorship.
FIELD	Consider queries of the form “find documents authored by William Shakespeare in 1601, containing the phrase alas poor Yorick”. Query processing then consists as usual of postings intersections, except that we may merge postings from standard inverted as well as <i>parametric indexes</i> . There is one parametric index for each field (say, date of creation); it allows us to select only the documents matching a date specified in the query. Figure 6.1 illustrates the user’s view of such a parametric search. Some of the fields may assume ordered values, such as dates; in the example query above, the year 1601 is one such field value. The search engine may support querying ranges on such ordered values; to this end, a structure like a B-tree may be used for the field’s dictionary.
PARAMETRIC INDEX	<i>Zones</i> are similar to fields, except the contents of a zone can be arbitrary free text. Whereas a field may take on a relatively small set of values, a zone can be thought of as an arbitrary, unbounded amount of text. For instance, document titles and abstracts are generally treated as zones. We may build a separate inverted index for each zone of a document, to support queries such as “find documents with merchant in the title and william in the author list and the phrase gentle rain in the body”. This has the effect of building an index that looks like Figure 6.2. Whereas the dictionary for a parametric index comes from a fixed vocabulary (the set of languages, or the set of dates), the dictionary for a zone index must structure whatever vocabulary stems from the text of that zone.
ZONE	In fact, we can reduce the size of the dictionary by encoding the zone in which a term occurs in the postings. In Figure 6.3 for instance, we show how occurrences of william in the title and author zones of various documents are encoded. Such an encoding is useful when the size of the dictionary is a concern (because we require the dictionary to fit in main memory). But there is another important reason why the encoding of Figure 6.3 is useful: the efficient computation of scores using a technique we will call <i>weighted zone scoring</i> .
WEIGHTED ZONE SCORING	

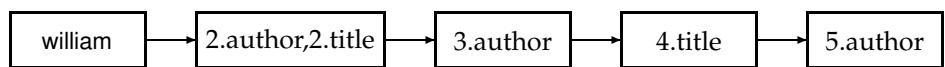
Bibliographic Search

Search category	Value
Author	Example: Widom, J or Garcia-Molina
Title	Also a part of the title possible
Date of publication	Example: 1997 or <1997 or >1997 limits the search to the documents appeared in, before and after 1997 respectively
Language	Language the document was written in English
Project	ANY
Type	ANY
Subject group	ANY
Sorted by	Date of publication
Start bibliographic search	
Find document via ID	

► **Figure 6.1** Parametric search. In this example we have a collection with fields allowing us to select publications by zones such as Author and fields such as Language.



► **Figure 6.2** Basic zone index ; zones are encoded as extensions of dictionary entries.



► **Figure 6.3** Zone index in which the zone is encoded in the postings rather than the dictionary.

6.1.1 Weighted zone scoring

Thus far in Section 6.1 we have focused on retrieving documents based on Boolean queries on fields and zones. We now turn to a second application of zones and fields.

Given a Boolean query q and a document d , weighted zone scoring assigns to the pair (q, d) a score in the interval $[0, 1]$, by computing a linear combination of *zone scores*, where each zone of the document contributes a Boolean value. More specifically, consider a set of documents each of which has ℓ zones. Let $g_1, \dots, g_\ell \in [0, 1]$ such that $\sum_{i=1}^{\ell} g_i = 1$. For $1 \leq i \leq \ell$, let s_i be the Boolean score denoting a match (or absence thereof) between q and the i th zone. For instance, the Boolean score from a zone could be 1 if all the query term(s) occur in that zone, and zero otherwise; indeed, it could be any Boolean function that maps the presence of query terms in a zone to 0, 1. Then, the weighted zone score is defined to be

$$(6.1) \quad \sum_{i=1}^{\ell} g_i s_i.$$

RANKED BOOLEAN
RETRIEVAL

Weighted zone scoring is sometimes referred to also as *ranked Boolean retrieval*.



Example 6.1: Consider the query *shakespeare* in a collection in which each document has three zones: *author*, *title* and *body*. The Boolean score function for a zone takes on the value 1 if the query term *shakespeare* is present in the zone, and zero otherwise. Weighted zone scoring in such a collection would require three weights g_1, g_2 and g_3 , respectively corresponding to the *author*, *title* and *body* zones. Suppose we set $g_1 = 0.2$, $g_2 = 0.3$ and $g_3 = 0.5$ (so that the three weights add up to 1); this corresponds to an application in which a match in the *author* zone is least important to the overall score, the *title* zone somewhat more, and the *body* contributes even more.

Thus if the term *shakespeare* were to appear in the *title* and *body* zones but not the *author* zone of a document, the score of this document would be 0.8.

How do we implement the computation of weighted zone scores? A simple approach would be to compute the score for each document in turn, adding in all the contributions from the various zones. However, we now show how we may compute weighted zone scores directly from inverted indexes. The algorithm of Figure 6.4 treats the case when the query q is a two-term query consisting of query terms q_1 and q_2 , and the Boolean function is AND: 1 if both query terms are present in a zone and 0 otherwise. Following the description of the algorithm, we describe the extension to more complex queries and Boolean functions.

The reader may have noticed the close similarity between this algorithm and that in Figure 1.6. Indeed, they represent the same postings traversal, except that instead of merely adding a document to the set of results for

```

ZONESCORE( $q_1, q_2$ )
1   float  $scores[N] = [0]$ 
2   constant  $g[\ell]$ 
3    $p_1 \leftarrow postings(q_1)$ 
4    $p_2 \leftarrow postings(q_2)$ 
5   //  $scores[]$  is an array with a score entry for each document, initialized to zero.
6   //  $p_1$  and  $p_2$  are initialized to point to the beginning of their respective postings.
7   // Assume  $g[]$  is initialized to the respective zone weights.
8   while  $p_1 \neq \text{NIL}$  and  $p_2 \neq \text{NIL}$ 
9     do if  $docID(p_1) = docID(p_2)$ 
10       then  $scores[docID(p_1)] \leftarrow \text{WEIGHTEDZONE}(p_1, p_2, g)$ 
11        $p_1 \leftarrow next(p_1)$ 
12        $p_2 \leftarrow next(p_2)$ 
13     else if  $docID(p_1) < docID(p_2)$ 
14       then  $p_1 \leftarrow next(p_1)$ 
15       else  $p_2 \leftarrow next(p_2)$ 
16   return  $scores$ 

```

► **Figure 6.4** Algorithm for computing the weighted zone score from two postings lists. Function WEIGHTEDZONE (not shown here) is assumed to compute the inner loop of Equation 6.1.

ACCUMULATOR

a Boolean AND query, we now compute a score for each such document. Some literature refers to the array $scores[]$ above as a set of *accumulators*. The reason for this will be clear as we consider more complex Boolean functions than the AND; thus we may assign a non-zero score to a document even if it does not contain all query terms.

MACHINE-LEARNED
RELEVANCE

6.1.2 Learning weights

How do we determine the weights g_i for weighted zone scoring? These weights could be specified by an expert (or, in principle, the user); but increasingly, these weights are “learned” using training examples that have been judged editorially. This latter methodology falls under a general class of approaches to scoring and ranking in information retrieval, known as *machine-learned relevance*. We provide a brief introduction to this topic here because weighted zone scoring presents a clean setting for introducing it; a complete development demands an understanding of machine learning and is deferred to Chapter 15.

1. We are provided with a set of *training examples*, each of which is a tuple consisting of a query q and a document d , together with a relevance

judgment for d on q . In the simplest form, each relevance judgments is either *Relevant* or *Non-relevant*. More sophisticated implementations of the methodology make use of more nuanced judgments.

2. The weights g_i are then “learned” from these examples, in order that the learned scores approximate the relevance judgments in the training examples.

For weighted zone scoring, the process may be viewed as learning a linear function of the Boolean match scores contributed by the various zones. The expensive component of this methodology is the labor-intensive assembly of user-generated relevance judgments from which to learn the weights, especially in a collection that changes frequently (such as the Web). We now detail a simple example that illustrates how we can reduce the problem of learning the weights g_i to a simple optimization problem.

We now consider a simple case of weighted zone scoring, where each document has a *title* zone and a *body* zone. Given a query q and a document d , we use the given Boolean match function to compute Boolean variables $s_T(d, q)$ and $s_B(d, q)$, depending on whether the title (respectively, body) zone of d matches query q . For instance, the algorithm in Figure 6.4 uses an AND of the query terms for this Boolean function. We will compute a score between 0 and 1 for each (document, query) pair using $s_T(d, q)$ and $s_B(d, q)$ by using a constant $g \in [0, 1]$, as follows:

$$(6.2) \quad \text{score}(d, q) = g \cdot s_T(d, q) + (1 - g)s_B(d, q).$$

We now describe how to determine the constant g from a set of *training examples*, each of which is a triple of the form $\Phi_j = (d_j, q_j, r(d_j, q_j))$. In each training example, a given training document d_j and a given training query q_j are assessed by a human editor who delivers a relevance judgment $r(d_j, q_j)$ that is either *Relevant* or *Non-relevant*. This is illustrated in Figure 6.5, where seven training examples are shown.

For each training example Φ_j we have Boolean values $s_T(d_j, q_j)$ and $s_B(d_j, q_j)$ that we use to compute a score from (6.2)

$$(6.3) \quad \text{score}(d_j, q_j) = g \cdot s_T(d_j, q_j) + (1 - g)s_B(d_j, q_j).$$

We now compare this computed score to the human relevance judgment for the same document-query pair (d_j, q_j) ; to this end, we will quantize each *Relevant* judgment as a 1 and each *Non-relevant* judgment as a 0. Suppose that we define the error of the scoring function with weight g as

$$\varepsilon(g, \Phi_j) = (r(d_j, q_j) - \text{score}(d_j, q_j))^2,$$

Example	DocID	Query	s_T	s_B	Judgment
Φ_1	37	linux	1	1	Relevant
Φ_2	37	penguin	0	1	Non-relevant
Φ_3	238	system	0	1	Relevant
Φ_4	238	penguin	0	0	Non-relevant
Φ_5	1741	kernel	1	1	Relevant
Φ_6	2094	driver	0	1	Relevant
Φ_7	3191	driver	1	0	Non-relevant

► **Figure 6.5** An illustration of training examples.

s_T	s_B	Score
0	0	0
0	1	$1 - g$
1	0	g
1	1	1

► **Figure 6.6** The four possible combinations of s_T and s_B .

where we have quantized the editorial relevance judgment $r(d_j, q_j)$ to 0 or 1. Then, the total error of a set of training examples is given by

$$(6.4) \quad \sum_j \varepsilon(g, \Phi_j).$$

The problem of learning the constant g from the given training examples then reduces to picking the value of g that minimizes the total error in (6.4).

Picking the best value of g in (6.4) in the formulation of Section 6.1.3 reduces to the problem of minimizing a quadratic function of g over the interval $[0, 1]$. This reduction is detailed in Section 6.1.3.



6.1.3 The optimal weight g

We begin by noting that for any training example Φ_j for which $s_T(d_j, q_j) = 0$ and $s_B(d_j, q_j) = 1$, the score computed by Equation (6.2) is $1 - g$. In similar fashion, we may write down the score computed by Equation (6.2) for the three other possible combinations of $s_T(d_j, q_j)$ and $s_B(d_j, q_j)$; this is summarized in Figure 6.6.

Let n_{01r} (respectively, n_{01n}) denote the number of training examples for which $s_T(d_j, q_j) = 0$ and $s_B(d_j, q_j) = 1$ and the editorial judgment is *Relevant* (respectively, *Non-relevant*). Then the contribution to the total error in Equation (6.4) from training examples for which $s_T(d_j, q_j) = 0$ and $s_B(d_j, q_j) = 1$

is

$$[1 - (1 - g)]^2 n_{01r} + [0 - (1 - g)]^2 n_{01n}.$$

By writing in similar fashion the error contributions from training examples of the other three combinations of values for $s_T(d_j, q_j)$ and $s_B(d_j, q_j)$ (and extending the notation in the obvious manner), the total error corresponding to Equation (6.4) is

$$(6.5) \quad (n_{01r} + n_{10n})g^2 + (n_{10r} + n_{01n})(1 - g)^2 + n_{00r} + n_{11n}.$$

By differentiating Equation (6.5) with respect to g and setting the result to zero, it follows that the optimal value of g is

$$(6.6) \quad \frac{n_{10r} + n_{01n}}{n_{10r} + n_{10n} + n_{01r} + n_{01n}}.$$



Exercise 6.1

When using weighted zone scoring, is it necessary for all zones to use the same Boolean match function?

Exercise 6.2

In Example 6.1 above with weights $g_1 = 0.2$, $g_2 = 0.31$ and $g_3 = 0.49$, what are all the distinct score values a document may get?

Exercise 6.3

Rewrite the algorithm in Figure 6.4 to the case of more than two query terms.

Exercise 6.4

Write pseudocode for the function WeightedZone for the case of two postings lists in Figure 6.4.

Exercise 6.5

Apply Equation 6.6 to the sample training set in Figure 6.5 to estimate the best value of g for this sample.

Exercise 6.6

For the value of g estimated in Exercise 6.5, compute the weighted zone score for each (query, document) example. How do these scores relate to the relevance judgments in Figure 6.5 (quantized to 0/1)?

Exercise 6.7

Why does the expression for g in (6.6) not involve training examples in which $s_T(d_t, q_t)$ and $s_B(d_t, q_t)$ have the same value?

6.2 Term frequency and weighting

Thus far, scoring has hinged on whether or not a query term is present in a zone within a document. We take the next logical step: a document or zone that mentions a query term more often has more to do with that query and therefore should receive a higher score. To motivate this, we recall the notion of a free text query introduced in Section 1.4: a query in which the terms of the query are typed freeform into the search interface, without any connecting search operators (such as Boolean operators). This query style, which is extremely popular on the web, views the query as simply a set of words. A plausible scoring mechanism then is to compute a score that is the sum, over the query terms, of the match scores between each query term and the document.

Towards this end, we assign to each term in a document a *weight* for that term, that depends on the number of occurrences of the term in the document. We would like to compute a score between a query term t and a document d , based on the weight of t in d . The simplest approach is to assign the weight to be equal to the number of occurrences of term t in document d . This weighting scheme is referred to as *term frequency* and is denoted $\text{tf}_{t,d}$, with the subscripts denoting the term and the document in order.

For a document d , the set of weights determined by the tf weights above (or indeed any weighting function that maps the number of occurrences of t in d to a positive real value) may be viewed as a quantitative digest of that document. In this view of a document, known in the literature as the *bag of words model*, the exact ordering of the terms in a document is ignored but the number of occurrences of each term is material (in contrast to Boolean retrieval). We only retain information on the number of occurrences of each term. Thus, the document “Mary is quicker than John” is, in this view, identical to the document “John is quicker than Mary”. Nevertheless, it seems intuitive that two documents with similar bag of words representations are similar in content. We will develop this intuition further in Section 6.3.

Before doing so we first study the question: are all words in a document equally important? Clearly not; in Section 2.2.2 (page 27) we looked at the idea of *stop words* – words that we decide not to index at all, and therefore do not contribute in any way to retrieval and scoring.

6.2.1 Inverse document frequency

Raw term frequency as above suffers from a critical problem: all terms are considered equally important when it comes to assessing relevancy on a query. In fact certain terms have little or no discriminating power in determining relevance. For instance, a collection of documents on the auto industry is likely to have the term *auto* in almost every document. To this

Word	cf	df
try	10422	8760
insurance	10440	3997

► **Figure 6.7** Collection frequency (cf) and document frequency (df) behave differently, as in this example from the Reuters collection.

end, we introduce a mechanism for attenuating the effect of terms that occur too often in the collection to be meaningful for relevance determination. An immediate idea is to scale down the term weights of terms with high *collection frequency*, defined to be the total number of occurrences of a term in the collection. The idea would be to reduce the tf weight of a term by a factor that grows with its collection frequency.

Instead, it is more commonplace to use for this purpose the *document frequency* df_t , defined to be the number of documents in the collection that contain a term t . This is because in trying to discriminate between documents for the purpose of scoring it is better to use a document-level statistic (such as the number of documents containing a term) than to use a collection-wide statistic for the term. The reason to prefer df to cf is illustrated in Figure 6.7, where a simple example shows that collection frequency (cf) and document frequency (df) can behave rather differently. In particular, the cf values for both try and insurance are roughly equal, but their df values differ significantly. Intuitively, we want the few documents that contain insurance to get a higher boost for a query on insurance than the many documents containing try get from a query on try.

How is the document frequency df of a term used to scale its weight? Denoting as usual the total number of documents in a collection by N , we define the *inverse document frequency* (idf) of a term t as follows:

$$(6.7) \quad idf_t = \log \frac{N}{df_t}.$$

Thus the idf of a rare term is high, whereas the idf of a frequent term is likely to be low. Figure 6.8 gives an example of idf's in the Reuters collection of 806,791 documents; in this example logarithms are to the base 10. In fact, as we will see in Exercise 6.12, the precise base of the logarithm is not material to ranking. We will give on page 227 a justification of the particular form in Equation (6.7).

6.2.2 Tf-idf weighting

We now combine the definitions of term frequency and inverse document frequency, to produce a composite weight for each term in each document.

term	df_t	idf_t
car	18,165	1.65
auto	6723	2.08
insurance	19,241	1.62
best	25,235	1.5

► **Figure 6.8** Example of idf values. Here we give the idf's of terms with various frequencies in the Reuters collection of 806,791 documents.

TF-IDF The *tf-idf* weighting scheme assigns to term t a weight in document d given by

$$(6.8) \quad \text{tf-idf}_{t,d} = \text{tf}_{t,d} \times \text{idf}_t.$$

In other words, $\text{tf-idf}_{t,d}$ assigns to term t a weight in document d that is

1. highest when t occurs many times within a small number of documents (thus lending high discriminating power to those documents);
2. lower when the term occurs fewer times in a document, or occurs in many documents (thus offering a less pronounced relevance signal);
3. lowest when the term occurs in virtually all documents.

DOCUMENT VECTOR

At this point, we may view each document as a *vector* with one component corresponding to each term in the dictionary, together with a weight for each component that is given by (6.8). For dictionary terms that do not occur in a document, this weight is zero. This vector form will prove to be crucial to scoring and ranking; we will develop these ideas in Section 6.3. As a first step, we introduce the *overlap score measure*: the score of a document d is the sum, over all query terms, of the number of times each of the query terms occurs in d . We can refine this idea so that we add up not the number of occurrences of each query term t in d , but instead the tf-idf weight of each term in d .

$$(6.9) \quad \text{Score}(q, d) = \sum_{t \in q} \text{tf-idf}_{t,d}.$$

In Section 6.3 we will develop a more rigorous form of Equation (6.9).



Exercise 6.8

Why is the idf of a term always finite?

Exercise 6.9

What is the idf of a term that occurs in every document? Compare this with the use of stop word lists.

	Doc1	Doc2	Doc3
car	27	4	24
auto	3	33	0
insurance	0	33	29
best	14	0	17

► **Figure 6.9** Table of tf values for Exercise 6.10.

Exercise 6.10

Consider the table of term frequencies for 3 documents denoted Doc1, Doc2, Doc3 in Figure 6.9. Compute the tf-idf weights for the terms car, auto, insurance, best, for each document, using the idf values from Figure 6.8.

Exercise 6.11

Can the tf-idf weight of a term in a document exceed 1?

Exercise 6.12

How does the base of the logarithm in (6.7) affect the score calculation in (6.9)? How does the base of the logarithm affect the relative scores of two documents on a given query?

Exercise 6.13

If the logarithm in (6.7) is computed base 2, suggest a simple approximation to the idf of a term.

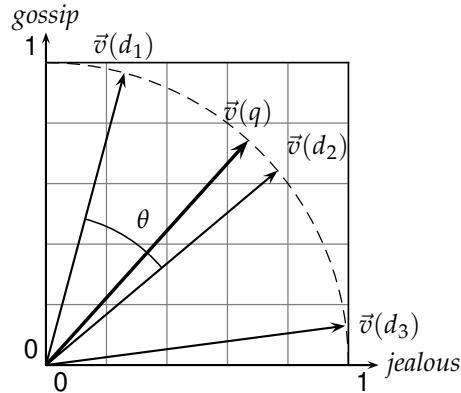
6.3 The vector space model for scoring

VECTOR SPACE MODEL

In Section 6.2 (page 117) we developed the notion of a document vector that captures the relative importance of the terms in a document. The representation of a set of documents as vectors in a common vector space is known as the *vector space model* and is fundamental to a host of information retrieval operations ranging from scoring documents on a query, document classification and document clustering. We first develop the basic ideas underlying vector space scoring; a pivotal step in this development is the view (Section 6.3.2) of queries as vectors in the same vector space as the document collection.

6.3.1 Dot products

We denote by $\vec{V}(d)$ the vector derived from document d , with one component in the vector for each dictionary term. Unless otherwise specified, the reader may assume that the components are computed using the tf-idf weighting scheme, although the particular weighting scheme is immaterial to the discussion that follows. The set of documents in a collection then may be viewed as a set of vectors in a vector space, in which there is one axis for



► **Figure 6.10** Cosine similarity illustrated. $\text{sim}(d_1, d_2) = \cos \theta$.

each term. This representation loses the relative ordering of the terms in each document; recall our example from Section 6.2 (page 117), where we pointed out that the documents *Mary is quicker than John* and *John is quicker than Mary* are identical in such a *bag of words* representation.

How do we quantify the similarity between two documents in this vector space? A first attempt might consider the magnitude of the vector difference between two document vectors. This measure suffers from a drawback: two documents with very similar content can have a significant vector difference simply because one is much longer than the other. Thus the relative distributions of terms may be identical in the two documents, but the absolute term frequencies of one may be far larger.

To compensate for the effect of document length, the standard way of quantifying the similarity between two documents d_1 and d_2 is to compute the *cosine similarity* of their vector representations $\vec{V}(d_1)$ and $\vec{V}(d_2)$

$$(6.10) \quad \text{sim}(d_1, d_2) = \frac{\vec{V}(d_1) \cdot \vec{V}(d_2)}{|\vec{V}(d_1)| |\vec{V}(d_2)|},$$

DOT PRODUCT
EUCLIDEAN LENGTH
LENGTH-NORMALIZATION

where the numerator represents the *dot product* (also known as the *inner product*) of the vectors $\vec{V}(d_1)$ and $\vec{V}(d_2)$, while the denominator is the product of their *Euclidean lengths*. The dot product $\vec{x} \cdot \vec{y}$ of two vectors is defined as $\sum_{i=1}^M x_i y_i$. Let $\vec{V}(d)$ denote the document vector for d , with M components $\vec{V}_1(d) \dots \vec{V}_M(d)$. The Euclidean length of d is defined to be $\sqrt{\sum_{i=1}^M \vec{V}_i^2(d)}$. The effect of the denominator of Equation (6.10) is thus to *length-normalize* the vectors $\vec{V}(d_1)$ and $\vec{V}(d_2)$ to unit vectors $\vec{v}(d_1) = \vec{V}(d_1)/|\vec{V}(d_1)|$ and

	Doc1	Doc2	Doc3
car	0.88	0.09	0.58
auto	0.10	0.71	0
insurance	0	0.71	0.70
best	0.46	0	0.41

► **Figure 6.11** Euclidean normalized tf values for documents in Figure 6.9.

term	SaS	PaP	WH
affection	115	58	20
jealous	10	7	11
gossip	2	0	6

► **Figure 6.12** Term frequencies in three novels. The novels are Austen’s *Sense and Sensibility*, *Pride and Prejudice* and Brontë’s *Wuthering Heights*.

$\vec{v}(d_2) = \vec{V}(d_2) / |\vec{V}(d_2)|$. We can then rewrite (6.10) as

$$(6.11) \quad \text{sim}(d_1, d_2) = \vec{v}(d_1) \cdot \vec{v}(d_2).$$

 **Example 6.2:** Consider the documents in Figure 6.9. We now apply Euclidean normalization to the tf values from the table, for each of the three documents in the table. The quantity $\sqrt{\sum_{i=1}^M V_i^2(d)}$ has the values 30.56, 46.84 and 41.30 respectively for Doc1, Doc2 and Doc3. The resulting Euclidean normalized tf values for these documents are shown in Figure 6.11.

Thus, (6.11) can be viewed as the dot product of the normalized versions of the two document vectors. This measure is the cosine of the angle θ between the two vectors, shown in Figure 6.10. What use is the similarity measure $\text{sim}(d_1, d_2)$? Given a document d (potentially one of the d_i in the collection), consider searching for the documents in the collection most similar to d . Such a search is useful in a system where a user may identify a document and seek others like it – a feature available in the results lists of search engines as a *more like this* feature. We reduce the problem of finding the document(s) most similar to d to that of finding the d_i with the highest dot products (sim values) $\vec{v}(d) \cdot \vec{v}(d_i)$. We could do this by computing the dot products between $\vec{v}(d)$ and each of $\vec{v}(d_1), \dots, \vec{v}(d_N)$, then picking off the highest resulting sim values.

 **Example 6.3:** Figure 6.12 shows the number of occurrences of three terms (affection, jealous and gossip) in each of the following three novels: Jane Austen’s *Sense and Sensibility* (SaS) and *Pride and Prejudice* (PaP) and Emily Brontë’s *Wuthering Heights* (WH).

term	SaS	PaP	WH
affection	0.996	0.993	0.847
jealous	0.087	0.120	0.466
gossip	0.017	0	0.254

► **Figure 6.13** Term vectors for the three novels of Figure 6.12. These are based on raw term frequency only and are normalized as if these were the only terms in the collection. (Since affection and jealous occur in all three documents, their tf-idf weight would be 0 in most formulations.)

Of course, there are many other terms occurring in each of these novels. In this example we represent each of these novels as a unit vector in three dimensions, corresponding to these three terms (only); we use raw term frequencies here, with no idf multiplier. The resulting weights are as shown in Figure 6.13.

Now consider the cosine similarities between pairs of the resulting three-dimensional vectors. A simple computation shows that $\text{sim}(\vec{v}(\text{SAS}), \vec{v}(\text{PAP}))$ is 0.999, whereas $\text{sim}(\vec{v}(\text{SAS}), \vec{v}(\text{WH}))$ is 0.888; thus, the two books authored by Austen (SaS and PaP) are considerably closer to each other than to Bronte's *Wuthering Heights*. In fact, the similarity between the first two is almost perfect (when restricted to the three terms we consider). Here we have considered tf weights, but we could of course use other term weight functions.

TERM-DOCUMENT
MATRIX

Viewing a collection of N documents as a collection of vectors leads to a natural view of a collection as a *term-document matrix*: this is an $M \times N$ matrix whose rows represent the M terms (dimensions) of the N columns, each of which corresponds to a document. As always, the terms being indexed could be stemmed before indexing; for instance, jealous and jealousy would under stemming be considered as a single dimension. This matrix view will prove to be useful in Chapter 18.

6.3.2 Queries as vectors

There is a far more compelling reason to represent documents as vectors: we can also view a *query* as a vector. Consider the query $q = \text{jealous gossip}$. This query turns into the unit vector $\vec{v}(q) = (0, 0.707, 0.707)$ on the three coordinates of Figures 6.12 and 6.13. The key idea now: to assign to each document d a score equal to the dot product

$$\vec{v}(q) \cdot \vec{v}(d).$$

In the example of Figure 6.13, *Wuthering Heights* is the top-scoring document for this query with a score of 0.509, with *Pride and Prejudice* a distant second with a score of 0.085, and *Sense and Sensibility* last with a score of 0.074. This simple example is somewhat misleading: the number of dimen-

sions in practice will be far larger than three: it will equal the vocabulary size M .

To summarize, by viewing a query as a “bag of words”, we are able to treat it as a very short document. As a consequence, we can use the cosine similarity between the query vector and a document vector as a measure of the score of the document for that query. The resulting scores can then be used to select the top-scoring documents for a query. Thus we have

$$(6.12) \quad \text{score}(q, d) = \frac{\vec{V}(q) \cdot \vec{V}(d)}{|\vec{V}(q)| |\vec{V}(d)|}.$$

A document may have a high cosine score for a query even if it does not contain all query terms. Note that the preceding discussion does not hinge on any specific weighting of terms in the document vector, although for the present we may think of them as either tf or tf-idf weights. In fact, a number of weighting schemes are possible for query as well as document vectors, as illustrated in Example 6.4 and developed further in Section 6.4.

Computing the cosine similarities between the query vector and each document vector in the collection, sorting the resulting scores and selecting the top K documents can be expensive — a single similarity computation can entail a dot product in tens of thousands of dimensions, demanding tens of thousands of arithmetic operations. In Section 7.1 we study how to use an inverted index for this purpose, followed by a series of heuristics for improving on this.

 **Example 6.4:** We now consider the query best car insurance on a fictitious collection with $N = 1,000,000$ documents where the document frequencies of auto, best, car and insurance are respectively 5000, 50000, 10000 and 1000.

term	query				document			product
	tf	df	idf	$w_{t,q}$	tf	wf	$w_{t,d}$	
auto	0	5000	2.3	0	1	1	0.41	0
best	1	50000	1.3	1.3	0	0	0	0
car	1	10000	2.0	2.0	1	1	0.41	0.82
insurance	1	1000	3.0	3.0	2	2	0.82	2.46

In this example the weight of a term in the query is simply the idf (and zero for a term not in the query, such as auto); this is reflected in the column header $w_{t,q}$ (the entry for auto is zero because the query does not contain the term auto). For documents, we use tf weighting with no use of idf but with Euclidean normalization. The former is shown under the column headed wf, while the latter is shown under the column headed $w_{t,d}$. Invoking (6.9) now gives a net score of $0 + 0 + 0.82 + 2.46 = 3.28$.

6.3.3 Computing vector scores

In a typical setting we have a collection of documents each represented by a vector, a free text query represented by a vector, and a positive integer K . We

```

COSINESCORE( $q$ )
1 float Scores[ $N$ ] = 0
2 Initialize Length[ $N$ ]
3 for each query term  $t$ 
4   do calculate  $w_{t,q}$  and fetch postings list for  $t$ 
5     for each pair( $d, tf_{t,d}$ ) in postings list
6       do Scores[ $d$ ] +=  $wf_{t,d} \times w_{t,q}$ 
7   Read the array Length[ $d$ ]
8   for each  $d$ 
9     do Scores[ $d$ ] = Scores[ $d$ ] / Length[ $d$ ]
10  return Top  $K$  components of Scores[]

```

► **Figure 6.14** The basic algorithm for computing vector space scores.

seek the K documents of the collection with the highest vector space scores on the given query. We now initiate the study of determining the K documents with the highest vector space scores for a query. Typically, we seek these K top documents in ordered by decreasing score; for instance many search engines use $K = 10$ to retrieve and rank-order the first page of the ten best results. Here we give the basic algorithm for this computation; we develop a fuller treatment of efficient techniques and approximations in Chapter 7.

Figure 6.14 gives the basic algorithm for computing vector space scores. The array Length holds the lengths (normalization factors) for each of the N documents, whereas the array Scores holds the scores for each of the documents. When the scores are finally computed in Step 9, all that remains in Step 10 is to pick off the K documents with the highest scores.

The outermost loop beginning Step 3 repeats the updating of Scores, iterating over each query term t in turn. In Step 5 we calculate the weight in the query vector for term t . Steps 6-8 update the score of each document by adding in the contribution from term t . This process of adding in contributions one query term at a time is sometimes known as *term-at-a-time* scoring or accumulation, and the N elements of the array Scores are therefore known as *accumulators*. For this purpose, it would appear necessary to store, with each postings entry, the weight $wf_{t,d}$ of term t in document d (we have thus far used either tf or $tf\text{-}idf$ for this weight, but leave open the possibility of other functions to be developed in Section 6.4). In fact this is wasteful, since storing this weight may require a floating point number. Two ideas help alleviate this space problem. First, if we are using inverse document frequency, we need not precompute idf_t ; it suffices to store N/df_t at the head of the postings for t . Second, we store the term frequency $tf_{t,d}$ for each postings entry. Finally, Step 12 extracts the top K scores – this requires a priority queue

TERM-AT-A-TIME

ACCUMULATOR

DOCUMENT-AT-A-TIME

data structure, often implemented using a heap. Such a heap takes no more than $2N$ comparisons to construct, following which each of the K top scores can be extracted from the heap at a cost of $O(\log N)$ comparisons.

Note that the general algorithm of Figure 6.14 does not prescribe a specific implementation of how we traverse the postings lists of the various query terms; we may traverse them one term at a time as in the loop beginning at Step 3, or we could in fact traverse them concurrently as in Figure 1.6. In such a concurrent postings traversal we compute the scores of one document at a time, so that it is sometimes called *document-at-a-time* scoring. We will say more about this in Section 7.1.5.



Exercise 6.14

If we were to stem jealous and jealousy to a common stem before setting up the vector space, detail how the definitions of tf and idf should be modified.

Exercise 6.15

Recall the tf-idf weights computed in Exercise 6.10. Compute the Euclidean normalized document vectors for each of the documents, where each vector has four components, one for each of the four terms.

Exercise 6.16

Verify that the sum of the squares of the components of each of the document vectors in Exercise 6.15 is 1 (to within rounding error). Why is this the case?

Exercise 6.17

With term weights as computed in Exercise 6.15, rank the three documents by computed score for the query car insurance, for each of the following cases of term weighting in the query:

1. The weight of a term is 1 if present in the query, 0 otherwise.
2. Euclidean normalized idf.

6.4 Variant tf-idf functions

For assigning a weight for each term in each document, a number of alternatives to tf and tf-idf have been considered. We discuss some of the principal ones here; a more complete development is deferred to Chapter 11. We will summarize these alternatives in Section 6.4.3 (page 128).

6.4.1 Sublinear tf scaling

It seems unlikely that twenty occurrences of a term in a document truly carry twenty times the significance of a single occurrence. Accordingly, there has been considerable research into variants of term frequency that go beyond counting the number of occurrences of a term. A common modification is

to use instead the logarithm of the term frequency, which assigns a weight given by

$$(6.13) \quad \text{wf}_{t,d} = \begin{cases} 1 + \log \text{tf}_{t,d} & \text{if } \text{tf}_{t,d} > 0 \\ 0 & \text{otherwise} \end{cases}.$$

In this form, we may replace tf by some other function wf as in (6.13), to obtain:

$$(6.14) \quad \text{wf-idf}_{t,d} = \text{wf}_{t,d} \times \text{idf}_t.$$

Equation (6.9) can then be modified by replacing tf-idf by wf-idf as defined in (6.14).

6.4.2 Maximum tf normalization

One well-studied technique is to normalize the tf weights of all terms occurring in a document by the maximum tf in that document. For each document d , let $\text{tf}_{\max}(d) = \max_{\tau \in d} \text{tf}_{\tau,d}$, where τ ranges over all terms in d . Then, we compute a normalized term frequency for each term t in document d by

$$(6.15) \quad \text{ntf}_{t,d} = a + (1 - a) \frac{\text{tf}_{t,d}}{\text{tf}_{\max}(d)},$$

where a is a value between 0 and 1 and is generally set to 0.4, although some early work used the value 0.5. The term a in (6.15) is a *smoothing* term whose role is to damp the contribution of the second term – which may be viewed as a scaling down of tf by the largest tf value in d . We will encounter smoothing further in Chapter 13 when discussing classification; the basic idea is to avoid a large swing in $\text{ntf}_{t,d}$ from modest changes in $\text{tf}_{t,d}$ (say from 1 to 2). The main idea of maximum tf normalization is to mitigate the following anomaly: we observe higher term frequencies in longer documents, merely because longer documents tend to repeat the same words over and over again. To appreciate this, consider the following extreme example: supposed we were to take a document d and create a new document d' by simply appending a copy of d to itself. While d' should be no more relevant to any query than d is, the use of (6.9) would assign it twice as high a score as d . Replacing $\text{tf-idf}_{t,d}$ in (6.9) by $\text{ntf-idf}_{t,d}$ eliminates the anomaly in this example. Maximum tf normalization does suffer from the following issues:

1. The method is unstable in the following sense: a change in the stop word list can dramatically alter term weightings (and therefore ranking). Thus, it is hard to tune.
2. A document may contain an outlier term with an unusually large number of occurrences of that term, not representative of the content of that document.

Term frequency		Document frequency		Normalization	
n (natural)	$tf_{t,d}$	n (no)	1	n (none)	1
l (logarithm)	$1 + \log(tf_{t,d})$	t (idf)	$\log \frac{N}{df_t}$	c (cosine)	$\frac{1}{\sqrt{w_1^2 + w_2^2 + \dots + w_M^2}}$
a (augmented)	$0.5 + \frac{0.5 \times tf_{t,d}}{\max_t(tf_{t,d})}$	p (prob idf)	$\max\{0, \log \frac{N - df_t}{df_t}\}$	u (pivoted unique)	$1/u$ (Section 6.4.4)
b (boolean)	$\begin{cases} 1 & \text{if } tf_{t,d} > 0 \\ 0 & \text{otherwise} \end{cases}$			b (byte size)	$1/CharLength^\alpha, \alpha < 1$
L (log ave)	$\frac{1 + \log(tf_{t,d})}{1 + \log(\text{ave}_{t \in d}(tf_{t,d}))}$				

► **Figure 6.15** SMART notation for tf-idf variants. Here $CharLength$ is the number of characters in the document.

- More generally, a document in which the most frequent term appears roughly as often as many other terms should be treated differently from one with a more skewed distribution.

6.4.3 Document and query weighting schemes

Equation (6.12) is fundamental to information retrieval systems that use any form of vector space scoring. Variations from one vector space scoring method to another hinge on the specific choices of weights in the vectors $\vec{V}(d)$ and $\vec{V}(q)$. Figure 6.15 lists some of the principal weighting schemes in use for each of $\vec{V}(d)$ and $\vec{V}(q)$, together with a mnemonic for representing a specific combination of weights; this system of mnemonics is sometimes called SMART notation, following the authors of an early text retrieval system. The mnemonic for representing a combination of weights takes the form $ddd.qqq$ where the first triplet gives the term weighting of the document vector, while the second triplet gives the weighting in the query vector. The first letter in each triplet specifies the term frequency component of the weighting, the second the document frequency component, and the third the form of normalization used. It is quite common to apply different normalization functions to $\vec{V}(d)$ and $\vec{V}(q)$. For example, a very standard weighting scheme is *lnc.ltc*, where the document vector has log-weighted term frequency, no idf (for both effectiveness and efficiency reasons), and cosine normalization, while the query vector uses log-weighted term frequency, idf weighting, and cosine normalization.



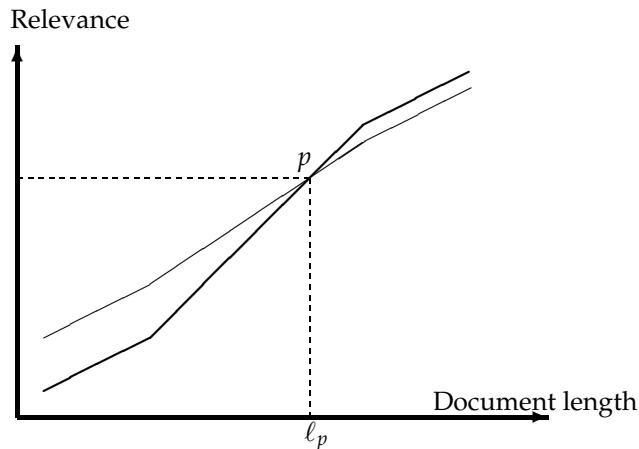
6.4.4 Pivoted normalized document length

In Section 6.3.1 we normalized each document vector by the Euclidean length of the vector, so that all document vectors turned into unit vectors. In doing so, we eliminated all information on the length of the original document; this masks some subtleties about longer documents. First, longer documents will – as a result of containing more terms – have higher tf values. Second, longer documents contain more distinct terms. These factors can conspire to raise the scores of longer documents, which (at least for some information needs) is unnatural. Longer documents can broadly be lumped into two categories: (1) *verbose* documents that essentially repeat the same content – in these, the length of the document does not alter the relative weights of different terms; (2) documents covering multiple different topics, in which the search terms probably match small segments of the document but not all of it – in this case, the relative weights of terms are quite different from a single short document that matches the query terms. Compensating for this phenomenon is a form of document length normalization that is independent of term and document frequencies. To this end, we introduce a form of normalizing the vector representations of documents in the collection, so that the resulting “normalized” documents are not necessarily of unit length. Then, when we compute the dot product score between a (unit) query vector and such a normalized document, the score is skewed to account for the effect of document length on relevance. This form of compensation for document length is known as *pivoted document length normalization*.

PIVOTED DOCUMENT
LENGTH
NORMALIZATION

Consider a document collection together with an ensemble of queries for that collection. Suppose that we were given, for each query q and for each document d , a Boolean judgment of whether or not d is relevant to the query q ; in Chapter 8 we will see how to procure such a set of relevance judgments for a query ensemble and a document collection. Given this set of relevance judgments, we may compute a *probability of relevance* as a function of document length, averaged over all queries in the ensemble. The resulting plot may look like the curve drawn in thick lines in Figure 6.16. To compute this curve, we bucket documents by length and compute the fraction of relevant documents in each bucket, then plot this fraction against the median document length of each bucket. (Thus even though the “curve” in Figure 6.16 appears to be continuous, it is in fact a histogram of discrete buckets of document length.)

On the other hand, the curve in thin lines shows what might happen with the same documents and query ensemble if we were to use relevance as prescribed by cosine normalization Equation (6.12) – thus, cosine normalization has a tendency to distort the computed relevance vis-à-vis the true relevance, at the expense of longer documents. The thin and thick curves crossover at a point p corresponding to document length ℓ_p , which we refer to as the *pivot*



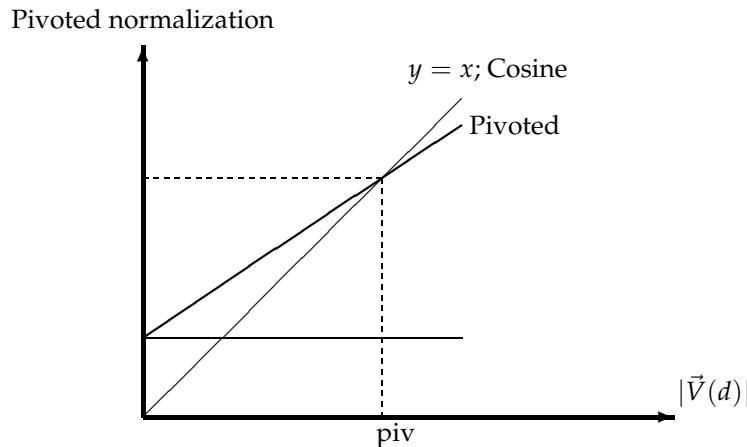
► **Figure 6.16** Pivoted document length normalization.

length; dashed lines mark this point on the x – and y – axes. The idea of pivoted document length normalization would then be to “rotate” the cosine normalization curve counter-clockwise about p so that it more closely matches thick line representing the relevance vs. document length curve. As mentioned at the beginning of this section, we do so by using in Equation (6.12) a normalization factor for each document vector $\vec{V}(d)$ that is not the Euclidean length of that vector, but instead one that is larger than the Euclidean length for documents of length less than ℓ_p , and smaller for longer documents.

To this end, we first note that the normalizing term for $\vec{V}(d)$ in the denominator of Equation (6.12) is its Euclidean length, denoted $|\vec{V}(d)|$. In the simplest implementation of pivoted document length normalization, we use a normalization factor in the denominator that is linear in $|\vec{V}(d)|$, but one of slope < 1 as in Figure 6.17. In this figure, the x – axis represents $|\vec{V}(d)|$, while the y – axis represents possible normalization factors we can use. The thin line $y = x$ depicts the use of cosine normalization. Notice the following aspects of the thick line representing pivoted length normalization:

1. It is linear in the document length and has the form

$$(6.16) \quad a|\vec{V}(d)| + (1 - a)\text{piv},$$



► **Figure 6.17** Implementing pivoted document length normalization by linear scaling.

where piv is the cosine normalization value at which the two curves intersect.

2. Its slope is $a < 1$ and (3) it crosses the $y = x$ line at piv .

It has been argued that in practice, Equation (6.16) is well approximated by

$$au_d + (1 - a)\text{piv},$$

where u_d is the number of unique terms in document d .

Of course, pivoted document length normalization is not appropriate for all applications. For instance, in a collection of answers to frequently asked questions (say, at a customer service website), relevance may have little to do with document length. In other cases the dependency may be more complex than can be accounted for by a simple linear pivoted normalization. In such cases, document length can be used as a feature in the machine learning based scoring approach of Section 6.1.2.

Exercise 6.18
EUCLIDEAN DISTANCE ?

One measure of the similarity of two vectors is the *Euclidean distance* (or L_2 distance) between them:

$$|\vec{x} - \vec{y}| = \sqrt{\sum_{i=1}^M (x_i - y_i)^2}$$

word	query					document			$q_i \cdot d_i$
	tf	wf	df	idf	$q_i = wf \cdot idf$	tf	wf	$d_i = \text{normalized wf}$	
digital				10,000					
video				100,000					
cameras				50,000					

► **Table 6.1** Cosine computation for Exercise 6.19.

Given a query q and documents d_1, d_2, \dots , we may rank the documents d_i in order of increasing Euclidean distance from q . Show that if q and the d_i are all normalized to unit vectors, then the rank ordering produced by Euclidean distance is identical to that produced by cosine similarities.

Exercise 6.19

Compute the vector space similarity between the query “digital cameras” and the document “digital cameras and video cameras” by filling out the empty columns in Table 6.1. Assume $N = 10,000,000$, logarithmic term weighting (wf columns) for query and document, idf weighting for the query only and cosine normalization for the document only. Treat and as a stop word. Enter term counts in the tf columns. What is the final similarity score?

Exercise 6.20

Show that for the query affection, the relative ordering of the scores of the three documents in Figure 6.13 is the reverse of the ordering of the scores for the query jealous gossip.

Exercise 6.21

In turning a query into a unit vector in Figure 6.13, we assigned equal weights to each of the query terms. What other principled approaches are plausible?

Exercise 6.22

Consider the case of a query term that is not in the set of M indexed terms; thus our standard construction of the query vector results in $\vec{V}(q)$ not being in the vector space created from the collection. How would one adapt the vector space representation to handle this case?

Exercise 6.23

Refer to the tf and idf values for four terms and three documents in Exercise 6.10. Compute the two top scoring documents on the query best car insurance for each of the following weighing schemes: (i) nnn . atc; (ii) ntc . atc.

Exercise 6.24

Suppose that the word coyote does not occur in the collection used in Exercises 6.10 and 6.23. How would one compute ntc . atc scores for the query coyote insurance?

6.5 References and further reading

Chapter 7 develops the computational aspects of vector space scoring. Luhn (1957; 1958) describes some of the earliest reported applications of term weighting. His paper dwells on the importance of medium frequency terms (terms that are neither too commonplace nor too rare) and may be thought of as anticipating tf-idf and related weighting schemes. Spärck Jones (1972) builds on this intuition through detailed experiments showing the use of inverse document frequency in term weighting. A series of extensions and theoretical justifications of idf are due to Salton and Buckley (1987) Robertson and Jones (1976), Croft and Harper (1979) and Papineni (2001). Robertson maintains a web page (<http://www.soi.city.ac.uk/~ser/idf.html>) containing the history of idf, including soft copies of early papers that predated electronic versions of journal article. Singhal et al. (1996a) develop pivoted document length normalization. Probabilistic language models (Chapter 11) develop weighting techniques that are more nuanced than tf-idf; the reader will find this development in Section 11.4.3.

We observed that by assigning a weight for each term in a document, a document may be viewed as a vector of term weights, one for each term in the collection. The SMART information retrieval system at Cornell (Salton 1971b) due to Salton and colleagues was perhaps the first to view a document as a vector of weights. The basic computation of cosine scores as described in Section 6.3.3 is due to Zobel and Moffat (2006). The two query evaluation strategies term-at-a-time and document-at-a-time are discussed by Turtle and Flood (1995).

The SMART notation for tf-idf term weighting schemes in Figure 6.15 is presented in (Salton and Buckley 1988, Singhal et al. 1995; 1996b). Not all versions of the notation are consistent; we most closely follow (Singhal et al. 1996b). A more detailed and exhaustive notation was developed in Moffat and Zobel (1998), considering a larger palette of schemes for term and document frequency weighting. Beyond the notation, Moffat and Zobel (1998) sought to set up a space of feasible weighting functions through which hill-climbing approaches could be used to begin with weighting schemes that performed well, then make local improvements to identify the best combinations. However, they report that such hill-climbing methods failed to lead to any conclusions on the best weighting schemes.

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7 Computing scores in a complete search system

Chapter 6 developed the theory underlying term weighting in documents for the purposes of scoring, leading up to vector space models and the basic cosine scoring algorithm of Section 6.3.3 (page 124). In this chapter we begin in Section 7.1 with heuristics for speeding up this computation; many of these heuristics achieve their speed at the risk of not finding quite the top K documents matching the query. Some of these heuristics generalize beyond cosine scoring. With Section 7.1 in place, we have essentially all the components needed for a complete search engine. We therefore take a step back from cosine scoring, to the more general problem of computing scores in a search engine. In Section 7.2 we outline a complete search engine, including indexes and structures to support not only cosine scoring but also more general ranking factors such as query term proximity. We describe how all of the various pieces fit together in Section 7.2.4. We conclude this chapter with Section 7.3, where we discuss how the vector space model for free text queries interacts with common query operators.

7.1 Efficient scoring and ranking

We begin by recapping the algorithm of Figure 6.14. For a query such as $q = \text{jealous gossip}$, two observations are immediate:

1. The unit vector $\vec{v}(q)$ has only two non-zero components.
2. In the absence of any weighting for query terms, these non-zero components are equal – in this case, both equal 0.707.

For the purpose of ranking the documents matching this query, we are really interested in the relative (rather than absolute) scores of the documents in the collection. To this end, it suffices to compute the cosine similarity from each document unit vector $\vec{v}(d)$ to $\vec{V}(q)$ (in which all non-zero components of the query vector are set to 1), rather than to the unit vector $\vec{v}(q)$. For any

```

FASTCOSINESCORE( $q$ )
1 float  $Scores[N] = 0$ 
2 for each  $d$ 
3 do Initialize  $Length[d]$  to the length of doc  $d$ 
4 for each query term  $t$ 
5 do calculate  $w_{t,q}$  and fetch postings list for  $t$ 
6   for each pair( $d, tf_{t,d}$ ) in postings list
7     do add  $wf_{t,d}$  to  $Scores[d]$ 
8 Read the array  $Length[d]$ 
9 for each  $d$ 
10 do Divide  $Scores[d]$  by  $Length[d]$ 
11 return Top  $K$  components of  $Scores[]$ 

```

► **Figure 7.1** A faster algorithm for vector space scores.

two documents d_1, d_2

$$(7.1) \quad \vec{V}(q) \cdot \vec{v}(d_1) > \vec{V}(q) \cdot \vec{v}(d_2) \Leftrightarrow \vec{v}(q) \cdot \vec{v}(d_1) > \vec{v}(q) \cdot \vec{v}(d_2).$$

For any document d , the cosine similarity $\vec{V}(q) \cdot \vec{v}(d)$ is the weighted sum, over all terms in the query q , of the weights of those terms in d . This in turn can be computed by a postings intersection exactly as in the algorithm of Figure 6.14, with line 8 altered since we take $w_{t,q}$ to be 1 so that the multiply-add in that step becomes just an addition; the result is shown in Figure 7.1. We walk through the postings in the inverted index for the terms in q , accumulating the total score for each document – very much as in processing a Boolean query, except we assign a positive score to each document that appears in any of the postings being traversed. As mentioned in Section 6.3.3 we maintain an idf value for each dictionary term and a tf value for each postings entry. This scheme computes a score for every document in the postings of any of the query terms; the total number of such documents may be considerably smaller than N .

Given these scores, the final step before presenting results to a user is to pick out the K highest-scoring documents. While one could sort the complete set of scores, a better approach is to use a heap to retrieve only the top K documents in order. Where J is the number of documents with non-zero cosine scores, constructing such a heap can be performed in $2J$ comparison steps, following which each of the K highest scoring documents can be “read off” the heap with $\log J$ comparison steps.

7.1.1 Inexact top K document retrieval

Thus far, we have focused on retrieving precisely the K highest-scoring documents for a query. We now consider schemes by which we produce K documents that are *likely* to be among the K highest scoring documents for a query. In doing so, we hope to dramatically lower the cost of computing the K documents we output, without materially altering the user's perceived relevance of the top K results. Consequently, in most applications it suffices to retrieve K documents whose scores are very close to those of the K best. In the sections that follow we detail schemes that retrieve K such documents while potentially avoiding computing scores for most of the N documents in the collection.

Such inexact top- K retrieval is not necessarily, from the user's perspective, a bad thing. The top K documents by the cosine measure are in any case not necessarily the K best for the query: cosine similarity is only a proxy for the user's perceived relevance. In Sections 7.1.2–7.1.6 below, we give heuristics using which we are likely to retrieve K documents with cosine scores close to those of the top K documents. The principal cost in computing the output stems from computing cosine similarities between the query and a large number of documents. Having a large number of documents in contention also increases the selection cost in the final stage of culling the top K documents from a heap. We now consider a series of ideas designed to eliminate a large number of documents without computing their cosine scores. The heuristics have the following two-step scheme:

1. Find a set A of documents that are contenders, where $K < |A| \ll N$. A does not necessarily contain the K top-scoring documents for the query, but is likely to have many documents with scores near those of the top K .
2. Return the K top-scoring documents in A .

From the descriptions of these ideas it will be clear that many of them require parameters to be tuned to the collection and application at hand; pointers to experience in setting these parameters may be found at the end of this chapter. It should also be noted that most of these heuristics are well-suited to free text queries, but not for Boolean or phrase queries.

7.1.2 Index elimination

For a multi-term query q , it is clear we only consider documents containing at least one of the query terms. We can take this a step further using additional heuristics:

1. We only consider documents containing terms whose idf exceeds a preset threshold. Thus, in the postings traversal, we only traverse the postings

for terms with high idf. This has a fairly significant benefit: the postings lists of low-idf terms are generally long; with these removed from contention, the set of documents for which we compute cosines is greatly reduced. One way of viewing this heuristic: low-idf terms are treated as stop words and do not contribute to scoring. For instance, on the query catcher in the rye, we only traverse the postings for catcher and rye. The cutoff threshold can of course be adapted in a query-dependent manner.

2. We only consider documents that contain many (and as a special case, all) of the query terms. This can be accomplished during the postings traversal; we only compute scores for documents containing all (or many) of the query terms. A danger of this scheme is that by requiring all (or even many) query terms to be present in a document before considering it for cosine computation, we may end up with fewer than K candidate documents in the output. This issue will be discussed further in Section 7.2.1.

7.1.3 Champion lists

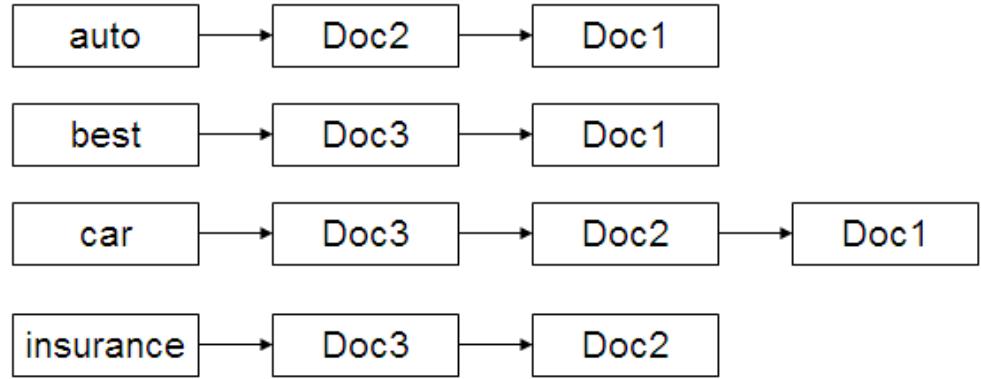
The idea of *champion lists* (sometimes also called *fancy lists* or *top docs*) is to precompute, for each term t in the dictionary, the set of the r documents with the highest weights for t ; the value of r is chosen in advance. For tf-idf weighting, these would be the r documents with the highest tf values for term t . We call this set of r documents the *champion list* for term t .

Now, given a query q we create a set A as follows: we take the union of the champion lists for each of the terms comprising q . We now restrict cosine computation to only the documents in A . A critical parameter in this scheme is the value r , which is highly application dependent. Intuitively, r should be large compared with K , especially if we use any form of the index elimination described in Section 7.1.2. One issue here is that the value r is set at the time of index construction, whereas K is application dependent and may not be available until the query is received; as a result we may (as in the case of index elimination) find ourselves with a set A that has fewer than K documents. There is no reason to have the same value of r for all terms in the dictionary; it could for instance be set to be higher for rarer terms.

7.1.4 Static quality scores and ordering

STATIC QUALITY SCORES

We now further develop the idea of champion lists, in the somewhat more general setting of *static quality scores*. In many search engines, we have available a measure of quality $g(d)$ for each document d that is query-independent and thus *static*. This quality measure may be viewed as a number between zero and one. For instance, in the context of news stories on the web, $g(d)$ may be derived from the number of favorable reviews of the story by web



► **Figure 7.2** A static quality-ordered index. In this example we assume that Doc1, Doc2 and Doc3 respectively have static quality scores $g(1) = 0.25, g(2) = 0.5, g(3) = 1$.

surfers. Section 4.6 (page 80) provides further discussion on this topic, as does Chapter 21 in the context of web search.

The net score for a document d is some combination of $g(d)$ together with the query-dependent score induced (say) by (6.12). The precise combination may be determined by the learning methods of Section 6.1.2, to be developed further in Section 15.4.1; but for the purposes of our exposition here, let us consider a simple sum:

$$(7.2) \quad \text{net-score}(q, d) = g(d) + \frac{\vec{V}(q) \cdot \vec{V}(d)}{|\vec{V}(q)| |\vec{V}(d)|}.$$

In this simple form, the static quality $g(d)$ and the query-dependent score from (6.10) have equal contributions, assuming each is between 0 and 1. Other relative weightings are possible; the effectiveness of our heuristics will depend on the specific relative weighting.

First, consider ordering the documents in the postings list for each term by decreasing value of $g(d)$. This allows us to perform the postings intersection algorithm of Figure 1.6. In order to perform the intersection by a single pass through the postings of each query term, the algorithm of Figure 1.6 relied on the postings being ordered by document IDs. But in fact, we only required that all postings be ordered by a single common ordering; here we rely on the $g(d)$ values to provide this common ordering. This is illustrated in Figure 7.2, where the postings are ordered in decreasing order of $g(d)$.

The first idea is a direct extension of champion lists: for a well-chosen value r , we maintain for each term t a *global champion list* of the r documents

with the highest values for $g(d) + \text{tf-idf}_{t,d}$. The list itself is, like all the postings lists considered so far, sorted by a common order (either by document IDs or by static quality). Then at query time, we only compute the net scores (7.2) for documents in the union of these global champion lists. Intuitively, this has the effect of focusing on documents likely to have large net scores.

We conclude the discussion of global champion lists with one further idea. We maintain for each term t two postings lists consisting of disjoint sets of documents, each sorted by $g(d)$ values. The first list, which we call *high*, contains the m documents with the highest tf values for t . The second list, which we call *low*, contains all other documents containing t . When processing a query, we first scan only the high lists of the query terms, computing net scores for any document on the high lists of all (or more than a certain number of) query terms. If we obtain scores for K documents in the process, we terminate. If not, we continue the scanning into the low lists, scoring documents in these postings lists. This idea is developed further in Section 7.2.1.

7.1.5 Impact ordering

In all the postings lists described thus far, we order the documents consistently by some common ordering: typically by document ID but in Section 7.1.4 by static quality scores. As noted at the end of Section 6.3.3, such a common ordering supports the concurrent traversal of all of the query terms' postings lists, computing the score for each document as we encounter it. Computing scores in this manner is sometimes referred to as document-at-a-time scoring. We will now introduce a technique for inexact top- K retrieval in which the postings are not all ordered by a common ordering, thereby precluding such a concurrent traversal. We will therefore require scores to be "accumulated" one term at a time as in the scheme of Figure 6.14, so that we have term-at-a-time scoring.

The idea is to order the documents d in the postings list of term t by decreasing order of $\text{tf}_{t,d}$. Thus, the ordering of documents will vary from one postings list to another, and we cannot compute scores by a concurrent traversal of the postings lists of all query terms. Given postings lists ordered by decreasing order of $\text{tf}_{t,d}$, two ideas have been found to significantly lower the number of documents for which we accumulate scores: (1) when traversing the postings list for a query term t , we stop after considering a prefix of the postings list – either after a fixed number of documents r have been seen, or after the value of $\text{tf}_{t,d}$ has dropped below a threshold; (2) when accumulating scores in the outer loop of Figure 6.14, we consider the query terms in decreasing order of idf, so that the query terms likely to contribute the most to the final scores are considered first. This latter idea too can be adaptive at the time of processing a query: as we get to query terms with lower idf, we can determine whether to proceed based on the changes in

document scores from processing the previous query term. If these changes are minimal, we may omit accumulation from the remaining query terms, or alternatively process shorter prefixes of their postings lists.

These ideas form a common generalization of the methods introduced in Sections 7.1.2–7.1.4. We may also implement a version of static ordering in which each postings list is ordered by an additive combination of static and query-dependent scores. We would again lose the consistency of ordering across postings, thereby having to process query terms one at time accumulating scores for all documents as we go along. Depending on the particular scoring function, the postings list for a document may be ordered by other quantities than term frequency; under this more general setting, this idea is known as impact ordering.

7.1.6 Cluster pruning

In *cluster pruning* we have a preprocessing step during which we cluster the document vectors. Then at query time, we consider only documents in a small number of clusters as candidates for which we compute cosine scores. Specifically, the preprocessing step is as follows:

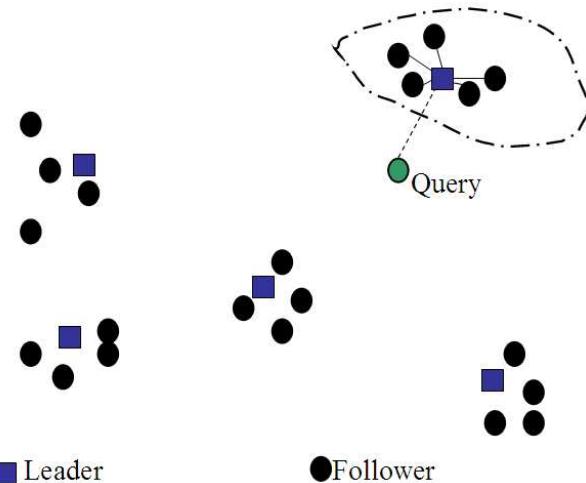
1. Pick \sqrt{N} documents at random from the collection. Call these *leaders*.
2. For each document that is not a leader, we compute its nearest leader.

We refer to documents that are not leaders as *followers*. Intuitively, in the partition of the followers induced by the use of \sqrt{N} randomly chosen leaders, the expected number of followers for each leader is $\approx N/\sqrt{N} = \sqrt{N}$. Next, query processing proceeds as follows:

1. Given a query q , find the leader L that is closest to q . This entails computing cosine similarities from q to each of the \sqrt{N} leaders.
2. The candidate set A consists of L together with its followers. We compute the cosine scores for all documents in this candidate set.

The use of randomly chosen leaders for clustering is fast and likely to reflect the distribution of the document vectors in the vector space: a region of the vector space that is dense in documents is likely to produce multiple leaders and thus a finer partition into sub-regions. This is illustrated in Figure 7.3.

Variations of cluster pruning introduce additional parameters b_1 and b_2 , both of which are positive integers. In the pre-processing step we attach each follower to its b_1 closest leaders, rather than a single closest leader. At query time we consider the b_2 leaders closest to the query q . Clearly, the basic scheme above corresponds to the case $b_1 = b_2 = 1$. Further, increasing b_1 or



► **Figure 7.3** Cluster pruning.

b_2 increases the likelihood of finding K documents that are more likely to be in the set of true top-scoring K documents, at the expense of more computation. We reiterate this approach when describing clustering in Chapter 16 (page 354).



Exercise 7.1

We suggested above (Figure 7.2) that the postings for static quality ordering be in decreasing order of $g(d)$. Why do we use the decreasing rather than the increasing order?

Exercise 7.2

When discussing champion lists, we simply used the r documents with the largest tf values to create the champion list for t . But when considering global champion lists, we used idf as well, identifying documents with the largest values of $g(d) + \text{tf-idf}_{t,d}$. Why do we differentiate between these two cases?

Exercise 7.3

If we were to only have one-term queries, explain why the use of global champion lists with $r = K$ suffices for identifying the K highest scoring documents. What is a simple modification to this idea if we were to only have s -term queries for any fixed integer $s > 1$?

Exercise 7.4

Explain how the common global ordering by $g(d)$ values in all high and low lists helps make the score computation efficient.

Exercise 7.5

Consider again the data of Exercise 6.23 with $nnn.\text{atc}$ for the query-dependent scoring. Suppose that we were given static quality scores of 1 for Doc1 and 2 for Doc2. Determine under Equation (7.2) what ranges of static quality score for Doc3 result in it being the first, second or third result for the query best car insurance.

Exercise 7.6

Sketch the frequency-ordered postings for the data in Figure 6.9.

Exercise 7.7

Let the static quality scores for Doc1, Doc2 and Doc3 in Figure 6.11 be respectively 0.25, 0.5 and 1. Sketch the postings for impact ordering when each postings list is ordered by the sum of the static quality score and the Euclidean normalized tf values in Figure 6.11.

Exercise 7.8

The nearest-neighbor problem in the plane is the following: given a set of N data points on the plane, we preprocess them into some data structure such that, given a query point Q , we seek the point in N that is closest to Q in Euclidean distance. Clearly cluster pruning can be used as an approach to the nearest-neighbor problem in the plane, if we wished to avoid computing the distance from Q to every one of the query points. Devise a simple example on the plane so that with two leaders, the answer returned by cluster pruning is incorrect (it is not the data point closest to Q).

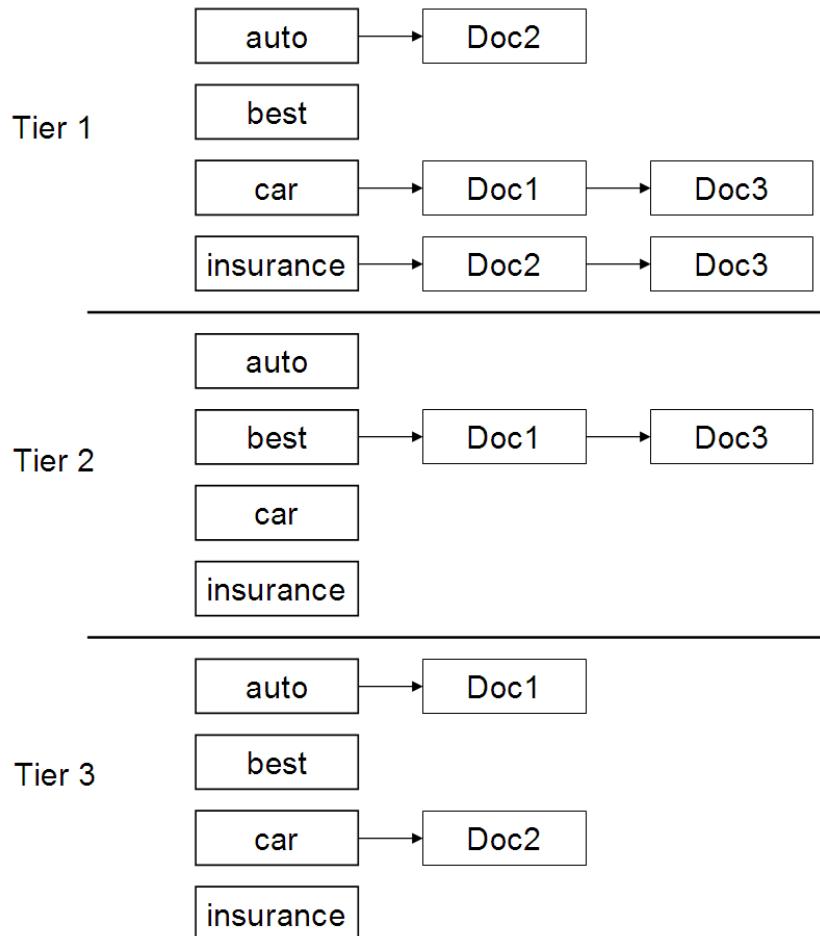
7.2 Components of an information retrieval system

In this section we combine the ideas developed so far to describe a rudimentary search system that retrieves and scores documents. We first develop further ideas for scoring, beyond vector spaces. Following this, we will put together all of these elements to outline a complete system. Because we consider a complete system, we do not restrict ourselves to vector space retrieval in this section. Indeed, our complete system will have provisions for vector space as well as other query operators and forms of retrieval. In Section 7.3 we will return to how vector space queries interact with other query operators.

7.2.1 Tiered indexes

TIERED INDEXES

We mentioned in Section 7.1.2 that when using heuristics such as index elimination for inexact top- K retrieval, we may occasionally find ourselves with a set A of contenders that has fewer than K documents. A common solution to this issue is the user of *tiered indexes*, which may be viewed as a generalization of champion lists. We illustrate this idea in Figure 7.4, where we represent the documents and terms of Figure 6.9. In this example we set a tf threshold of 20 for tier 1 and 10 for tier 2, meaning that the tier 1 index only has postings entries with tf values exceeding 20, while the tier 2 index only



► **Figure 7.4** Tiered indexes. If we fail to get K results from tier 1, query processing “falls back” to tier 2, and so on. Within each tier, postings are ordered by document ID.

has postings entries with tf values exceeding 10. In this example we have chosen to order the postings entries within a tier by document ID.

7.2.2 Query-term proximity

Especially for free text queries on the web (Chapter 19), users prefer a document in which most or all of the query terms appear close to each other,

because this is evidence that the document has text focused on their query intent. Consider a query with two or more query terms, t_1, t_2, \dots, t_k . Let ω be the width of the smallest window in a document d that contains all the query terms, measured in the number of words in the window. For instance, if the document were to simply consist of the sentence The quality of mercy is not strained, the smallest window for the query strained mercy would be 4. Intuitively, the smaller that ω is, the better that d matches the query. In cases where the document does not contain all of the query terms, we can set ω to be some enormous number. We could also consider variants in which only words that are not stop words are considered in computing ω . Such proximity-weighted scoring functions are a departure from pure cosine similarity and closer to the “soft conjunctive” semantics that Google and other web search engines evidently use.

PROXIMITY WEIGHTING

How can we design such a *proximity-weighted* scoring function to depend on ω ? The simplest answer relies on a “hand coding” technique we introduce below in Section 7.2.3. A more scalable approach goes back to Section 6.1.2 – we treat the integer ω as yet another feature in the scoring function, whose importance is assigned by machine learning, as will be developed further in Section 15.4.1.

7.2.3 Designing parsing and scoring functions

Common search interfaces, particularly for consumer-facing search applications on the web, tend to mask query operators from the end user. The intent is to hide the complexity of these operators from the largely non-technical audience for such applications, inviting *free text queries*. Given such interfaces, how should a search equipped with indexes for various retrieval operators treat a query such as rising interest rates? More generally, given the various factors we have studied that could affect the score of a document, how should we combine these features?

The answer of course depends on the user population, the query distribution and the collection of documents. Typically, a *query parser* is used to translate the user-specified keywords into a query with various operators that is executed against the underlying indexes. Sometimes, this execution can entail multiple queries against the underlying indexes; for example, the query parser may issue a stream of queries:

1. Run the user-generated query string as a phrase query. Rank them by vector space scoring using as query the vector consisting of the 3 terms rising interest rates.
2. If fewer than ten documents contain the phrase rising interest rates, run the two 2-term phrase queries rising interest and interest rates; rank these using vector space scoring, as well.

EVIDENCE ACCUMULATION

3. If we still have fewer than ten results, run the vector space query consisting of the three individual query terms.

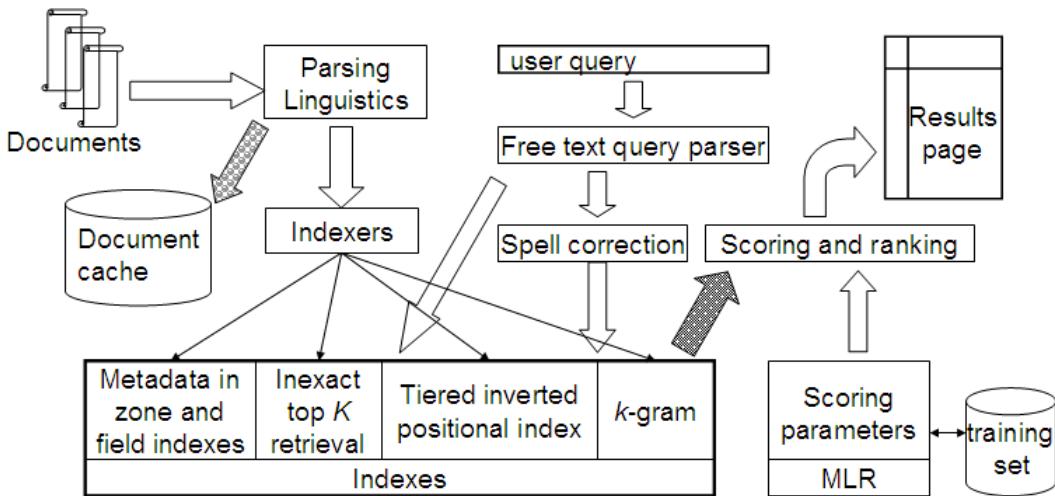
Each of these steps (if invoked) may yield a list of scored documents, for each of which we compute a score. This score must combine contributions from vector space scoring, static quality, proximity weighting and potentially other factors – particularly since a document may appear in the lists from multiple steps. This demands an aggregate scoring function that *accumulates evidence* of a document’s relevance from multiple sources. How do we devise a query parser and how do we devise the aggregate scoring function?

The answer depends on the setting. In many enterprise settings we have application builders who make use of a toolkit of available scoring operators, along with a query parsing layer, with which to manually configure the scoring function as well as the query parser. Such application builders make use of the available zones, metadata and knowledge of typical documents and queries to tune the parsing and scoring. In collections whose characteristics change infrequently (in an enterprise application, significant changes in collection and query characteristics typically happen with infrequent events such as the introduction of new document formats or document management systems, or a merger with another company). Web search on the other hand is faced with a constantly changing document collection with new characteristics being introduced all the time. It is also a setting in which the number of scoring factors can run into the hundreds, making hand-tuned scoring a difficult exercise. To address this, it is becoming increasingly common to use machine-learned scoring, extending the ideas we introduced in Section 6.1.2, as will be discussed further in Section 15.4.1.

7.2.4 Putting it all together

We have now studied all the components necessary for a basic search system that supports free text queries as well as Boolean, zone and field queries. We briefly review how the various pieces fit together into an overall system; this is depicted in Figure 7.5.

In this figure, documents stream in from the left for parsing and linguistic processing (language and format detection, tokenization and stemming). The resulting stream of tokens feeds into two modules. First, we retain a copy of each parsed document in a document cache. This will enable us to generate results snippets: snippets of text accompanying each document in the results list for a query. This snippet tries to give a succinct explanation to the user of why the document matches the query. The automatic generation of such snippets is the subject of Section 8.7. A second copy of the tokens is fed to a bank of indexers that create a bank of indexes including zone and field indexes that store the metadata for each document,



► **Figure 7.5** A complete search system. Data paths are shown primarily for a free text query.

(tiered) positional indexes, indexes for spelling correction and other tolerant retrieval, and structures for accelerating inexact top- K retrieval. A free text user query (top center) is sent down to the indexes both directly and through a module for generating spelling-correction candidates. As noted in Chapter 3 the latter may optionally be invoked only when the original query fails to retrieve enough results. Retrieved documents (dark arrow) are passed to a scoring module that computes scores based on machine-learned ranking (MLR), a technique that builds on Section 6.1.2 (to be further developed in Section 15.4.1) for scoring and ranking documents. Finally, these ranked documents are rendered as a results page.



Exercise 7.9

Explain how the postings intersection algorithm first introduced in Section 1.3 can be adapted to find the smallest integer ω that contains all query terms.

Exercise 7.10

Adapt this procedure to work when not all query terms are present in a document.

7.3 Vector space scoring and query operator interaction

We introduced the vector space model as a paradigm for free text queries. We conclude this chapter by discussing how the vector space scoring model

relates to the query operators we have studied in earlier chapters. The relationship should be viewed at two levels: in terms of the expressiveness of queries that a sophisticated user may pose, and in terms of the index that supports the evaluation of the various retrieval methods. In building a search engine, we may opt to support multiple query operators for an end user. In doing so we need to understand what components of the index can be shared for executing various query operators, as well as how to handle user queries that mix various query operators.

Vector space scoring supports so-called free text retrieval, in which a query is specified as a set of words without any query operators connecting them. It allows documents matching the query to be scored and thus ranked, unlike the Boolean, wildcard and phrase queries studied earlier. Classically, the interpretation of such free text queries was that at least one of the query terms be present in any retrieved document. However more recently, web search engines such as Google have popularized the notion that a set of terms typed into their query boxes (thus on the face of it, a free text query) carries the semantics of a conjunctive query that only retrieves documents containing all or most query terms.

Boolean retrieval

Clearly a vector space index can be used to answer Boolean queries, as long as the weight of a term t in the document vector for d is non-zero whenever t occurs in d . The reverse is not true, since a Boolean index does not by default maintain term weight information. There is no easy way of combining vector space and Boolean queries from a user's standpoint: vector space queries are fundamentally a form of *evidence accumulation*, where the presence of more query terms in a document adds to the score of a document. Boolean retrieval on the other hand, requires a user to specify a formula for *selecting* documents through the presence (or absence) of specific combinations of keywords, without inducing any relative ordering among them. Mathematically, it is in fact possible to invoke so-called p -norms to combine Boolean and vector space queries, but we know of no system that makes use of this fact.

Wildcard queries

Wildcard and vector space queries require different indexes, except at the basic level that both can be implemented using postings and a dictionary (e.g., a dictionary of trigrams for wildcard queries). If a search engine allows a user to specify a wildcard operator as part of a free text query (for instance, the query `rom* restaurant`), we may interpret the wildcard component of the query as spawning multiple terms in the vector space (in this example, `rome`

and roman would be two such terms) all of which are added to the query vector. The vector space query is then executed as usual, with matching documents being scored and ranked; thus a document containing both rome and roma is likely to be scored higher than another containing only one of them. The exact score ordering will of course depend on the relative weights of each term in matching documents.

Phrase queries

The representation of documents as vectors is fundamentally lossy: the relative order of terms in a document is lost in the encoding of a document as a vector. Even if we were to try and somehow treat every biword as a term (and thus an axis in the vector space), the weights on different axes not independent: for instance the phrase German shepherd gets encoded in the axis german shepherd, but immediately has a non-zero weight on the axes german and shepherd. Further, notions such as idf would have to be extended to such biwords. Thus an index built for vector space retrieval cannot, in general, be used for phrase queries. Moreover, there is no way of demanding a vector space score for a phrase query — we only know the relative weights of each term in a document.

On the query german shepherd, we could use vector space retrieval to identify documents heavy in these two terms, with no way of prescribing that they occur consecutively. Phrase retrieval, on the other hand, tells us of the existence of the phrase german shepherd in a document, without any indication of the relative frequency or weight of this phrase. While these two retrieval paradigms (phrase and vector space) consequently have different implementations in terms of indexes and retrieval algorithms, they can in some cases be combined usefully, as in the three-step example of query parsing in Section 7.2.3.

7.4 References and further reading

TOP DOCS

Heuristics for fast query processing with early termination are described by Anh et al. (2001), Garcia et al. (2004), Anh and Moffat (2006b), Persin et al. (1996). Cluster pruning is investigated by Singitham et al. (2004) and by Chierichetti et al. (2007); see also Section 16.6 (page 372). Champion lists are described in Persin (1994) and (under the name *top docs*) in Brown (1995), and further developed in Brin and Page (1998), Long and Suel (2003). While these heuristics are well-suited to free text queries that can be viewed as vectors, they complicate phrase queries; see Anh and Moffat (2006c) for an index structure that supports both weighted and Boolean/phrase searches. Carmel et al. (2001) Clarke et al. (2000) and Song et al. (2005) treat the use of query

term proximity in assessing relevance. Pioneering work on learning of ranking functions was done by Fuhr (1989), Fuhr and Pfeifer (1994), Cooper et al. (1994), Bartell (1994), Bartell et al. (1998) and by Cohen et al. (1998).

8 *Evaluation in information retrieval*

We have seen in the preceding chapters many alternatives in designing an IR system. How do we know which of these techniques are effective in which applications? Should we use stop lists? Should we stem? Should we use inverse document frequency weighting? Information retrieval has developed as a highly empirical discipline, requiring careful and thorough evaluation to demonstrate the superior performance of novel techniques on representative document collections.

In this chapter we begin with a discussion of measuring the effectiveness of IR systems (Section 8.1) and the test collections that are most often used for this purpose (Section 8.2). We then present the straightforward notion of relevant and nonrelevant documents and the formal evaluation methodology that has been developed for evaluating unranked retrieval results (Section 8.3). This includes explaining the kinds of evaluation measures that are standardly used for document retrieval and related tasks like text classification and why they are appropriate. We then extend these notions and develop further measures for evaluating ranked retrieval results (Section 8.4) and discuss developing reliable and informative test collections (Section 8.5).

We then step back to introduce the notion of user utility, and how it is approximated by the use of document relevance (Section 8.6). The key utility measure is user happiness. Speed of response and the size of the index are factors in user happiness. It seems reasonable to assume that relevance of results is the most important factor: blindingly fast, useless answers do not make a user happy. However, user perceptions do not always coincide with system designers' notions of quality. For example, user happiness commonly depends very strongly on user interface design issues, including the layout, clarity, and responsiveness of the user interface, which are independent of the quality of the results returned. We touch on other measures of the quality of a system, in particular the generation of high-quality result summary snippets, which strongly influence user utility, but are not measured in the basic relevance ranking paradigm (Section 8.7).

8.1 Information retrieval system evaluation

To measure ad hoc information retrieval effectiveness in the standard way, we need a test collection consisting of three things:

1. A document collection
2. A test suite of information needs, expressible as queries
3. A set of relevance judgments, standardly a binary assessment of either *relevant* or *nonrelevant* for each query-document pair.

RELEVANCE

GOLD STANDARD GROUND TRUTH

INFORMATION NEED

The standard approach to information retrieval system evaluation revolves around the notion of *relevant* and *nonrelevant* documents. With respect to a user information need, a document in the test collection is given a binary classification as either relevant or nonrelevant. This decision is referred to as the *gold standard* or *ground truth* judgment of relevance. The test document collection and suite of information needs have to be of a reasonable size: you need to average performance over fairly large test sets, as results are highly variable over different documents and information needs. As a rule of thumb, 50 information needs has usually been found to be a sufficient minimum.

Relevance is assessed relative to an information need, *not* a query. For example, an information need might be:

Information on whether drinking red wine is more effective at reducing your risk of heart attacks than white wine.

This might be translated into a query such as:

wine AND red AND white AND heart AND attack AND effective

A document is relevant if it addresses the stated information need, not because it just happens to contain all the words in the query. This distinction is often misunderstood in practice, because the information need is not overt. But, nevertheless, an information need is present. If a user types python into a web search engine, they might be wanting to know where they can purchase a pet python. Or they might be wanting information on the programming language Python. From a one word query, it is very difficult for a system to know what the information need is. But, nevertheless, the user has one, and can judge the returned results on the basis of their relevance to it. To evaluate a system, we require an overt expression of an information need, which can be used for judging returned documents as relevant or nonrelevant. At this point, we make a simplification: relevance can reasonably be thought of as a scale, with some documents highly relevant and others marginally so. But for the moment, we will use just a binary decision of relevance. We

DEVELOPMENT TEST
COLLECTION

discuss the reasons for using binary relevance judgments and alternatives in Section 8.5.1.

Many systems contain various weights (often known as parameters) that can be adjusted to tune system performance. It is wrong to report results on a test collection which were obtained by tuning these parameters to maximize performance on that collection. That is because such tuning overstates the expected performance of the system, because the weights will be set to maximize performance on one particular set of queries rather than for a random sample of queries. In such cases, the correct procedure is to have one or more *development test collections*, and to tune the parameters on the development test collection. The tester then runs the system with those weights on the test collection and reports the results on that collection as an unbiased estimate of performance.

8.2 Standard test collections

Here is a list of the most standard test collections and evaluation series. We focus particularly on test collections for ad hoc information retrieval system evaluation, but also mention a couple of similar test collections for text classification.

- | | |
|-----------|--------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| CRANFIELD | The <i>Cranfield</i> collection. This was the pioneering test collection in allowing precise quantitative measures of information retrieval effectiveness, but is nowadays too small for anything but the most elementary pilot experiments. Collected in the United Kingdom starting in the late 1950s, it contains 1398 abstracts of aerodynamics journal articles, a set of 225 queries, and exhaustive relevance judgments of all (query, document) pairs. |
| TREC | <i>Text Retrieval Conference (TREC)</i> . The U.S. National Institute of Standards and Technology (NIST) has run a large IR test bed evaluation series since 1992. Within this framework, there have been many tracks over a range of different test collections, but the best known test collections are the ones used for the TREC Ad Hoc track during the first 8 TREC evaluations between 1992 and 1999. In total, these test collections comprise 6 CDs containing 1.89 million documents (mainly, but not exclusively, newswire articles) and relevance judgments for 450 information needs, which are called <i>topics</i> and specified in detailed text passages. Individual test collections are defined over different subsets of this data. The early TRECs each consisted of 50 information needs, evaluated over different but overlapping sets of documents. TREC 6–8 provide 150 information needs over about 528,000 newswire and Foreign Broadcast Information Service articles. This is probably the best subcollection to use in future work, because it is the largest and the topics are more consistent. Because the test |

document collections are so large, there are no exhaustive relevance judgments. Rather, NIST assessors' relevance judgments are available only for the documents that were among the top k returned for some system which was entered in the TREC evaluation for which the information need was developed.

GOV2	In more recent years, NIST has done evaluations on larger document collections, including the 25 million page GOV2 web page collection. From the beginning, the NIST test document collections were orders of magnitude larger than anything available to researchers previously and GOV2 is now the largest Web collection easily available for research purposes. Nevertheless, the size of GOV2 is still more than 2 orders of magnitude smaller than the current size of the document collections indexed by the large web search companies.
NTCIR CROSS-LANGUAGE INFORMATION RETRIEVAL	NII Test Collections for IR Systems (NTCIR). The NTCIR project has built various test collections of similar sizes to the TREC collections, focusing on East Asian language and <i>cross-language information retrieval</i> , where queries are made in one language over a document collection containing documents in one or more other languages. See: http://research.nii.ac.jp/ntcir/data/data-en.html
CLEF	Cross Language Evaluation Forum (CLEF). This evaluation series has concentrated on European languages and cross-language information retrieval. See: http://www.clef-campaign.org/
REUTERS	Reuters-21578 and Reuters-RCV1. For text classification, the most used test collection has been the Reuters-21578 collection of 21578 newswire articles; see Chapter 13, page 279. More recently, Reuters released the much larger Reuters Corpus Volume 1 (RCV1), consisting of 806,791 documents; see Chapter 4, page 69. Its scale and rich annotation makes it a better basis for future research.
20 NEWSGROUPS	20 <i>Newsgroups</i> . This is another widely used text classification collection, collected by Ken Lang. It consists of 1000 articles from each of 20 Usenet newsgroups (the newsgroup name being regarded as the category). After the removal of duplicate articles, as it is usually used, it contains 18941 articles.

8.3 Evaluation of unranked retrieval sets

Given these ingredients, how is system effectiveness measured? The two most frequent and basic measures for information retrieval effectiveness are precision and recall. These are first defined for the simple case where an

IR system returns a set of documents for a query. We will see later how to extend these notions to ranked retrieval situations.

PRECISION *Precision (P)* is the fraction of retrieved documents that are relevant

$$(8.1) \quad \text{Precision} = \frac{\#\text{(relevant items retrieved)}}{\#\text{(retrieved items)}} = P(\text{relevant}|\text{retrieved})$$

RECALL *Recall (R)* is the fraction of relevant documents that are retrieved

$$(8.2) \quad \text{Recall} = \frac{\#\text{(relevant items retrieved)}}{\#\text{(relevant items)}} = P(\text{retrieved}|\text{relevant})$$

These notions can be made clear by examining the following contingency table:

	Relevant	Nonrelevant
Retrieved	true positives (tp)	false positives (fp)
Not retrieved	false negatives (fn)	true negatives (tn)

Then:

$$(8.4) \quad \begin{aligned} P &= tp/(tp + fp) \\ R &= tp/(tp + fn) \end{aligned}$$

ACCURACY An obvious alternative that may occur to the reader is to judge an information retrieval system by its *accuracy*, that is, the fraction of its classifications that are correct. In terms of the contingency table above, accuracy = $(tp + tn)/(tp + fp + fn + tn)$. This seems plausible, since there are two actual classes, relevant and nonrelevant, and an information retrieval system can be thought of as a two-class classifier which attempts to label them as such (it retrieves the subset of documents which it believes to be relevant). This is precisely the effectiveness measure often used for evaluating machine learning classification problems.

There is a good reason why accuracy is not an appropriate measure for information retrieval problems. In almost all circumstances, the data is extremely skewed: normally over 99.9% of the documents are in the nonrelevant category. A system tuned to maximize accuracy can appear to perform well by simply deeming all documents nonrelevant to all queries. Even if the system is quite good, trying to label some documents as relevant will almost always lead to a high rate of false positives. However, labeling all documents as nonrelevant is completely unsatisfying to an information retrieval system user. Users are always going to want to see some documents, and can be

assumed to have a certain tolerance for seeing some false positives providing that they get some useful information. The measures of precision and recall concentrate the evaluation on the return of true positives, asking what percentage of the relevant documents have been found and how many false positives have also been returned.

The advantage of having the two numbers for precision and recall is that one is more important than the other in many circumstances. Typical web surfers would like every result on the first page to be relevant (high precision) but have not the slightest interest in knowing let alone looking at every document that is relevant. In contrast, various professional searchers such as paralegals and intelligence analysts are very concerned with trying to get as high recall as possible, and will tolerate fairly low precision results in order to get it. Individuals searching their hard disks are also often interested in high recall searches. Nevertheless, the two quantities clearly trade off against one another: you can always get a recall of 1 (but very low precision) by retrieving all documents for all queries! Recall is a non-decreasing function of the number of documents retrieved. On the other hand, in a good system, precision usually decreases as the number of documents retrieved is increased. In general we want to get some amount of recall while tolerating only a certain percentage of false positives.

F MEASURE

A single measure that trades off precision versus recall is the *F measure*, which is the weighted harmonic mean of precision and recall:

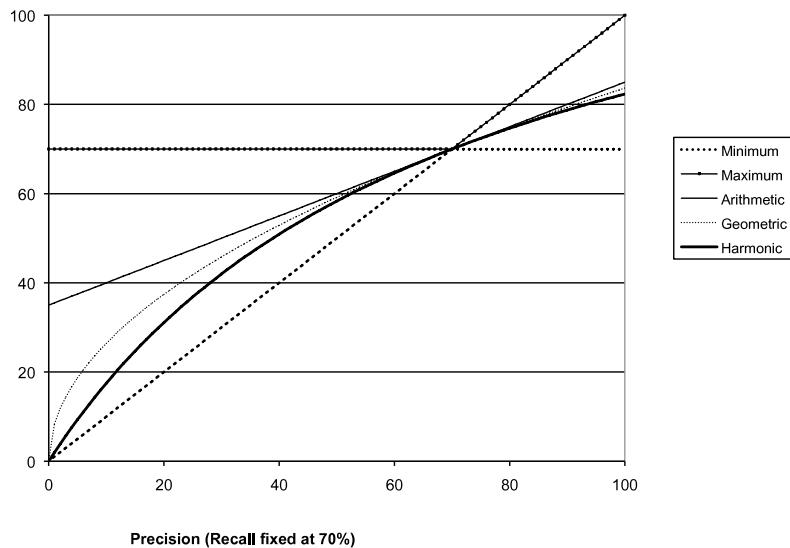
$$(8.5) \quad F = \frac{1}{\alpha \frac{1}{P} + (1 - \alpha) \frac{1}{R}} = \frac{(\beta^2 + 1)PR}{\beta^2 P + R} \quad \text{where} \quad \beta^2 = \frac{1 - \alpha}{\alpha}$$

where $\alpha \in [0, 1]$ and thus $\beta^2 \in [0, \infty]$. The default *balanced F measure* equally weights precision and recall, which means making $\alpha = 1/2$ or $\beta = 1$. It is commonly written as F_1 , which is short for $F_{\beta=1}$, even though the formulation in terms of α more transparently exhibits the F measure as a weighted harmonic mean. When using $\beta = 1$, the formula on the right simplifies to:

$$(8.6) \quad F_{\beta=1} = \frac{2PR}{P + R}$$

However, using an even weighting is not the only choice. Values of $\beta < 1$ emphasize precision, while values of $\beta > 1$ emphasize recall. For example, a value of $\beta = 3$ or $\beta = 5$ might be used if recall is to be emphasized. Recall, precision, and the F measure are inherently measures between 0 and 1, but they are also very commonly written as percentages, on a scale between 0 and 100.

Why do we use a harmonic mean rather than the simpler average (arithmetic mean)? Recall that we can always get 100% recall by just returning all documents, and therefore we can always get a 50% arithmetic mean by the



► **Figure 8.1** Graph comparing the harmonic mean to other means. The graph shows a slice through the calculation of various means of precision and recall for the fixed recall value of 70%. The harmonic mean is always less than either the arithmetic or geometric mean, and often quite close to the minimum of the two numbers. When the precision is also 70%, all the measures coincide.

same process. This strongly suggests that the arithmetic mean is an unsuitable measure to use. In contrast, if we assume that 1 document in 10,000 is relevant to the query, the harmonic mean score of this strategy is 0.02%. The harmonic mean is always less than or equal to the arithmetic mean and the geometric mean. When the values of two numbers differ greatly, the harmonic mean is closer to their minimum than to their arithmetic mean; see Figure 8.1.



Exercise 8.1

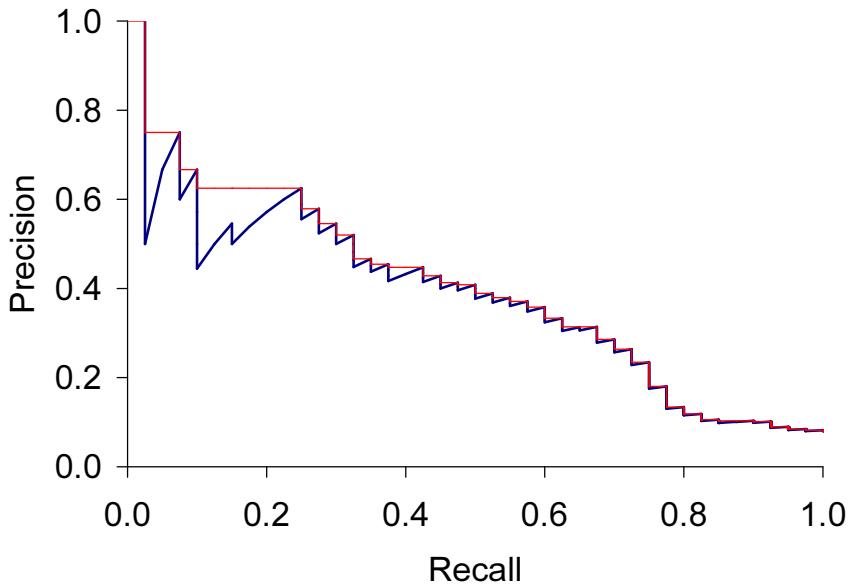
[\star]

An IR system returns 8 relevant documents, and 10 nonrelevant documents. There are a total of 20 relevant documents in the collection. What is the precision of the system on this search, and what is its recall?

Exercise 8.2

[\star]

The balanced F measure (a.k.a. F_1) is defined as the harmonic mean of precision and recall. What is the advantage of using the harmonic mean rather than “averaging” (using the arithmetic mean)?



► **Figure 8.2** Precision/recall graph.

Exercise 8.3

[**]

Derive the equivalence between the two formulas for F measure shown in Equation (8.5), given that $\alpha = 1/(\beta^2 + 1)$.

8.4 Evaluation of ranked retrieval results

Precision, recall, and the F measure are set-based measures. They are computed using unordered sets of documents. We need to extend these measures (or to define new measures) if we are to evaluate the ranked retrieval results that are now standard with search engines. In a ranked retrieval context, appropriate sets of retrieved documents are naturally given by the top k retrieved documents. For each such set, precision and recall values can be plotted to give a *precision-recall curve*, such as the one shown in Figure 8.2. Precision-recall curves have a distinctive saw-tooth shape: if the $(k+1)^{\text{th}}$ document retrieved is nonrelevant then recall is the same as for the top k documents, but precision has dropped. If it is relevant, then both precision and recall increase, and the curve jags up and to the right. It is often useful to remove these jiggles and the standard way to do this is with an interpolated precision: the *interpolated precision* p_{interp} at a certain recall level r is defined

PRECISION-RECALL
CURVE

INTERPOLATED
PRECISION

Recall	Interp. Precision
0.0	1.00
0.1	0.67
0.2	0.63
0.3	0.55
0.4	0.45
0.5	0.41
0.6	0.36
0.7	0.29
0.8	0.13
0.9	0.10
1.0	0.08

► **Table 8.1** Calculation of 11-point Interpolated Average Precision. This is for the precision-recall curve shown in Figure 8.2.

as the highest precision found for any recall level $r' \geq r$:

$$(8.7) \quad p_{\text{interp}}(r) = \max_{r' \geq r} p(r')$$

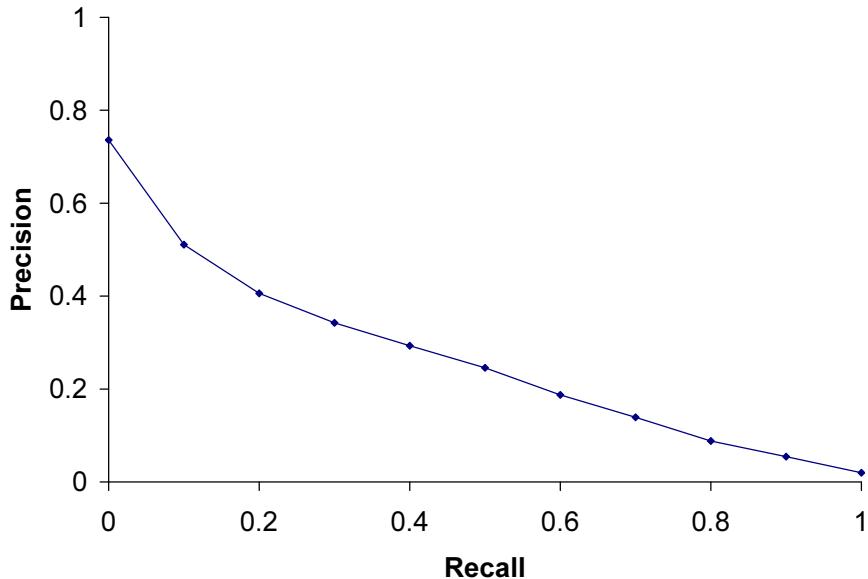
The justification is that almost anyone would be prepared to look at a few more documents if it would increase the percentage of the viewed set that were relevant (that is, if the precision of the larger set is higher). Interpolated precision is shown by a thinner line in Figure 8.2. With this definition, the interpolated precision at a recall of 0 is well-defined (Exercise 8.4).

Examining the entire precision-recall curve is very informative, but there is often a desire to boil this information down to a few numbers, or perhaps even a single number. The traditional way of doing this (used for instance in the first 8 TREC Ad Hoc evaluations) is the *11-point interpolated average precision*. For each information need, the interpolated precision is measured at the 11 recall levels of 0.0, 0.1, 0.2, ..., 1.0. For the precision-recall curve in Figure 8.2, these 11 values are shown in Table 8.1. For each recall level, we then calculate the arithmetic mean of the interpolated precision at that recall level for each information need in the test collection. A composite precision-recall curve showing 11 points can then be graphed. Figure 8.3 shows an example graph of such results from a representative good system at TREC 8.

In recent years, other measures have become more common. Most standard among the TREC community is *Mean Average Precision* (MAP), which provides a single-figure measure of quality across recall levels. Among evaluation measures, MAP has been shown to have especially good discrimination and stability. For a single information need, Average Precision is the

11-POINT
INTERPOLATED
AVERAGE PRECISION

MEAN AVERAGE
PRECISION



► **Figure 8.3** Averaged 11-point precision/recall graph across 50 queries for a representative TREC system. The Mean Average Precision for this system is 0.2553.

average of the precision value obtained for the set of top k documents existing after each relevant document is retrieved, and this value is then averaged over information needs. That is, if the set of relevant documents for an information need $q_j \in Q$ is $\{d_1, \dots, d_{m_j}\}$ and R_{jk} is the set of ranked retrieval results from the top result until you get to document d_k , then

$$(8.8) \quad \text{MAP}(Q) = \frac{1}{|Q|} \sum_{j=1}^{|Q|} \frac{1}{m_j} \sum_{k=1}^{m_j} \text{Precision}(R_{jk})$$

When a relevant document is not retrieved at all,¹ the precision value in the above equation is taken to be 0. For a single information need, the average precision approximates the area under the uninterpolated precision-recall curve, and so the MAP is roughly the average area under the precision-recall curve for a set of queries.

Using MAP, fixed recall levels are not chosen, and there is no interpolation. The MAP value for a test collection is the arithmetic mean of average

1. A system may not fully order all documents in the collection in response to a query or at any rate an evaluation exercise may be based on submitting only the top k results for each information need.

precision values for individual information needs. (This has the effect of weighting each information need equally in the final reported number, even if many documents are relevant to some queries whereas very few are relevant to other queries.) Calculated MAP scores normally vary widely across information needs when measured within a single system, for instance, between 0.1 and 0.7. Indeed, there is normally more agreement in MAP for an individual information need across systems than for MAP scores for different information needs for the same system. This means that a set of test information needs must be large and diverse enough to be representative of system effectiveness across different queries.

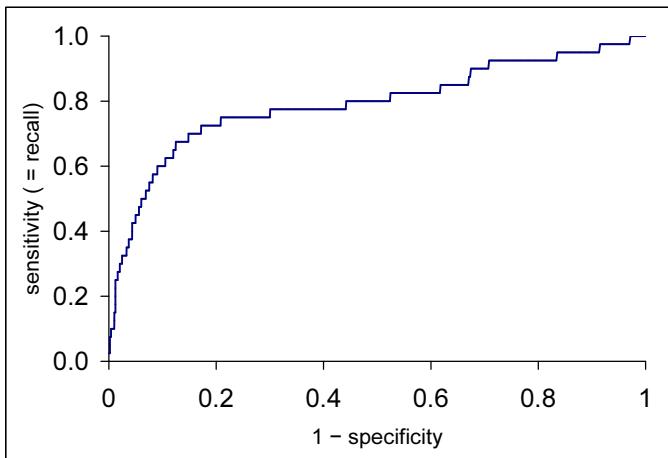
PRECISION AT k

The above measures factor in precision at all recall levels. For many prominent applications, particularly web search, this may not be germane to users. What matters is rather how many good results there are on the first page or the first three pages. This leads to measuring precision at fixed low levels of retrieved results, such as 10 or 30 documents. This is referred to as "Precision at k ", for example "Precision at 10". It has the advantage of not requiring any estimate of the size of the set of relevant documents but the disadvantages that it is the least stable of the commonly used evaluation measures and that it does not average well, since the total number of relevant documents for a query has a strong influence on precision at k .

R-PRECISION

An alternative, which alleviates this problem, is *R-precision*. It requires having a set of known relevant documents Rel , from which we calculate the precision of the top Rel documents returned. (The set Rel may be incomplete, such as when Rel is formed by creating relevance judgments for the pooled top k results of particular systems in a set of experiments.) R-precision adjusts for the size of the set of relevant documents: A perfect system could score 1 on this metric for each query, whereas, even a perfect system could only achieve a precision at 20 of 0.4 if there were only 8 documents in the collection relevant to an information need. Averaging this measure across queries thus makes more sense. This measure is harder to explain to naive users than Precision at k but easier to explain than MAP. If there are $|Rel|$ relevant documents for a query, we examine the top $|Rel|$ results of a system, and find that r are relevant, then by definition, not only is the precision (and hence R-precision) $r/|Rel|$, but the recall of this result set is also $r/|Rel|$. Thus, R-precision turns out to be identical to the *break-even point*, another measure which is sometimes used, defined in terms of this equality relationship holding. Like Precision at k , R-precision describes only one point on the precision-recall curve, rather than attempting to summarize effectiveness across the curve, and it is somewhat unclear why you should be interested in the break-even point rather than either the best point on the curve (the point with maximal F-measure) or a retrieval level of interest to a particular application (Precision at k). Nevertheless, R-precision turns out to be highly correlated with MAP empirically, despite measuring only a single point on

BREAK-EVEN POINT



► **Figure 8.4** The ROC curve corresponding to the precision-recall curve in Figure 8.2.

ROC CURVE
SENSITIVITY
SPECIFICITY
CUMULATIVE GAIN
NORMALIZED
DISCOUNTED
CUMULATIVE GAIN

the curve.

Another concept sometimes used in evaluation is an *ROC curve*. (“ROC” stands for “Receiver Operating Characteristics”, but knowing that doesn’t help most people.) An ROC curve plots the true positive rate or sensitivity against the false positive rate or $(1 - \text{specificity})$. Here, *sensitivity* is just another term for recall. The false positive rate is given by $fp/(fp + tn)$. Figure 8.4 shows the ROC curve corresponding to the precision-recall curve in Figure 8.2. An ROC curve always goes from the bottom left to the top right of the graph. For a good system, the graph climbs steeply on the left side. For unranked result sets, *specificity*, given by $tn/(fp + tn)$, was not seen as a very useful notion. Because the set of true negatives is always so large, its value would be almost 1 for all information needs (and, correspondingly, the value of the false positive rate would be almost 0). That is, the “interesting” part of Figure 8.2 is $0 < \text{recall} < 0.4$, a part which is compressed to a small corner of Figure 8.4. But an ROC curve could make sense when looking over the full retrieval spectrum, and it provides another way of looking at the data. In many fields, a common aggregate measure is to report the area under the ROC curve, which is the ROC analog of MAP. Precision-recall curves are sometimes loosely referred to as ROC curves. This is understandable, but not accurate.

A final approach that has seen increasing adoption, especially when employed with machine learning approaches to ranking (see Section 15.4, page 341) is measures of *cumulative gain*, and in particular *normalized discounted cumulative gain*.

NDCG *relative gain (NDCG)*. NDCG is designed for situations of non-binary notions of relevance (cf. Section 8.5.1). Like precision at k , it is evaluated over some number k of top search results. For a set of queries Q , let $R(j, d)$ be the relevance score assessors gave to document d for query j . Then,

$$(8.9) \quad \text{NDCG}(Q, k) = \frac{1}{|Q|} \sum_{j=1}^{|Q|} Z_{kj} \sum_{m=1}^k \frac{2^{R(j,m)} - 1}{\log_2(1 + m)},$$

where Z_{kj} is a normalization factor calculated to make it so that a perfect ranking's NDCG at k for query j is 1. For queries for which $k' < k$ documents are retrieved, the last summation is done up to k' .

**Exercise 8.4**

[★]

What are the possible values for interpolated precision at a recall level of 0?

Exercise 8.5

[★★]

Must there always be a break-even point between precision and recall? Either show there must be or give a counter-example.

Exercise 8.6

[★★]

What is the relationship between the value of F_1 and the break-even point?

Exercise 8.7

[★★]

DICE COEFFICIENT

The *Dice coefficient* of two sets is a measure of their intersection scaled by their size (giving a value in the range 0 to 1):

$$\text{Dice}(X, Y) = \frac{2|X \cap Y|}{|X| + |Y|}$$

Show that the balanced F-measure (F_1) is equal to the Dice coefficient of the retrieved and relevant document sets.

Exercise 8.8

[★]

Consider an information need for which there are 4 relevant documents in the collection. Contrast two systems run on this collection. Their top 10 results are judged for relevance as follows (the leftmost item is the top ranked search result):

System 1	R N R N N	N N N R R
System 2	N R N N R	R R N N N

- a. What is the MAP of each system? Which has a higher MAP?
- b. Does this result intuitively make sense? What does it say about what is important in getting a good MAP score?
- c. What is the R-precision of each system? (Does it rank the systems the same as MAP?)

Exercise 8.9

[**]

The following list of Rs and Ns represents relevant (R) and nonrelevant (N) returned documents in a ranked list of 20 documents retrieved in response to a query from a collection of 10,000 documents. The top of the ranked list (the document the system thinks is most likely to be relevant) is on the left of the list. This list shows 6 relevant documents. Assume that there are 8 relevant documents in total in the collection.

R R N N N N N N R N R N N N R N N N N R

- a. What is the precision of the system on the top 20?
- b. What is the F_1 on the top 20?
- c. What is the uninterpolated precision of the system at 25% recall?
- d. What is the interpolated precision at 33% recall?
- e. Assume that these 20 documents are the complete result set of the system. What is the MAP for the query?

Assume, now, instead, that the system returned the entire 10,000 documents in a ranked list, and these are the first 20 results returned.

- f. What is the largest possible MAP that this system could have?
- g. What is the smallest possible MAP that this system could have?
- h. In a set of experiments, only the top 20 results are evaluated by hand. The result in (e) is used to approximate the range (f)–(g). For this example, how large (in absolute terms) can the error for the MAP be by calculating (e) instead of (f) and (g) for this query?

8.5 Assessing relevance

To properly evaluate a system, your test information needs must be germane to the documents in the test document collection, and appropriate for predicted usage of the system. These information needs are best designed by domain experts. Using random combinations of query terms as an information need is generally not a good idea because typically they will not resemble the actual distribution of information needs.

Given information needs and documents, you need to collect relevance assessments. This is a time-consuming and expensive process involving human beings. For tiny collections like Cranfield, exhaustive judgments of relevance for each query and document pair were obtained. For large modern collections, it is usual for relevance to be assessed only for a subset of the documents for each query. The most standard approach is *pooling*, where relevance is assessed over a subset of the collection that is formed from the top k documents returned by a number of different IR systems (usually the ones to be evaluated), and perhaps other sources such as the results of Boolean keyword searches or documents found by expert searchers in an interactive process.

POOLING

		Judge 2 Relevance		
		Yes	No	Total
Judge 1 Relevance	Yes	300	20	320
	No	10	70	80
	Total	310	90	400

Observed proportion of the times the judges agreed

$$P(A) = (300 + 70)/400 = 370/400 = 0.925$$

Pooled marginals

$$P(\text{nonrelevant}) = (80 + 90)/(400 + 400) = 170/800 = 0.2125$$

$$P(\text{relevant}) = (320 + 310)/(400 + 400) = 630/800 = 0.7878$$

Probability that the two judges agreed by chance

$$P(E) = P(\text{nonrelevant})^2 + P(\text{relevant})^2 = 0.2125^2 + 0.7878^2 = 0.665$$

Kappa statistic

$$\kappa = (P(A) - P(E))/(1 - P(E)) = (0.925 - 0.665)/(1 - 0.665) = 0.776$$

► **Table 8.2** Calculating the kappa statistic.

A human is not a device that reliably reports a gold standard judgment of relevance of a document to a query. Rather, humans and their relevance judgments are quite idiosyncratic and variable. But this is not a problem to be solved: in the final analysis, the success of an IR system depends on how good it is at satisfying the needs of these idiosyncratic humans, one information need at a time.

Nevertheless, it is interesting to consider and measure how much agreement between judges there is on relevance judgments. In the social sciences, a common measure for agreement between judges is the *kappa statistic*. It is designed for categorical judgments and corrects a simple agreement rate for the rate of chance agreement.

KAPPA STATISTIC

MARGINAL

(8.10)

$$\kappa = \frac{P(A) - P(E)}{1 - P(E)}$$

where $P(A)$ is the proportion of the times the judges agreed, and $P(E)$ is the proportion of the times they would be expected to agree by chance. There are choices in how the latter is estimated: if we simply say we are making a two-class decision and assume nothing more, then the expected chance agreement rate is 0.5. However, normally the class distribution assigned is skewed, and it is usual to use *marginal* statistics to calculate expected agreement.² There are still two ways to do it depending on whether one pools

2. For a contingency table, as in Table 8.2, a marginal statistic is formed by summing a row or column. The marginal $a_{i,k} = \sum_j a_{ijk}$.

the marginal distribution across judges or uses the marginals for each judge separately; both forms have been used, but we present the pooled version because it is more conservative in the presence of systematic differences in assessments across judges. The calculations are shown in Table 8.2. The kappa value will be 1 if two judges always agree, 0 if they agree only at the rate given by chance, and negative if they are worse than random. If there are more than two judges, it is normal to calculate an average pairwise kappa value. As a rule of thumb, a kappa value above 0.8 is taken as good agreement, a kappa value between 0.67 and 0.8 is taken as fair agreement, and agreement below 0.67 is seen as data providing a dubious basis for an evaluation, though the precise cutoffs depend on the purposes for which the data will be used.

Interjudge agreement of relevance has been measured within the TREC evaluations and for medical IR collections. Using the above rules of thumb, the level of agreement normally falls in the range of “fair” (0.67–0.8). The fact that human agreement on a binary relevance judgment is quite modest is one reason for not requiring more fine-grained relevance labeling from the test set creator. To answer the question of whether IR evaluation results are valid despite the variation of individual assessors’ judgments, people have experimented with evaluations taking one or the other of two judges’ opinions as the gold standard. The choice can make a considerable *absolute* difference to reported scores, but has in general been found to have little impact on the *relative* effectiveness ranking of either different systems or variants of a single system which are being compared for effectiveness.

8.5.1 Critiques and justifications of the concept of relevance

The advantage of system evaluation, as enabled by the standard model of relevant and nonrelevant documents, is that we have a fixed setting in which we can vary IR systems and system parameters to carry out comparative experiments. Such formal testing is much less expensive and allows clearer diagnosis of the effect of changing system parameters than doing user studies of retrieval effectiveness. Indeed, once we have a formal measure that we have confidence in, we can proceed to optimize effectiveness by machine learning methods, rather than tuning parameters by hand. Of course, if the formal measure poorly describes what users actually want, doing this will not be effective in improving user satisfaction. Our perspective is that, in practice, the standard formal measures for IR evaluation, although a simplification, are good enough, and recent work in optimizing formal evaluation measures in IR has succeeded brilliantly. There are numerous examples of techniques developed in formal evaluation settings, which improve effectiveness in operational settings, such as the development of document length normalization methods within the context of TREC (Sections 6.4.4 and 11.4.3)

and machine learning methods for adjusting parameter weights in scoring (Section 6.1.2).

That is not to say that there are not problems latent within the abstractions used. The relevance of one document is treated as independent of the relevance of other documents in the collection. (This assumption is actually built into most retrieval systems – documents are scored against queries, not against each other – as well as being assumed in the evaluation methods.) Assessments are binary: there aren't any nuanced assessments of relevance. Relevance of a document to an information need is treated as an absolute, objective decision. But judgments of relevance are subjective, varying across people, as we discussed above. In practice, human assessors are also imperfect measuring instruments, susceptible to failures of understanding and attention. We also have to assume that users' information needs do not change as they start looking at retrieval results. Any results based on one collection are heavily skewed by the choice of collection, queries, and relevance judgment set: the results may not translate from one domain to another or to a different user population.

Some of these problems may be fixable. A number of recent evaluations, including INEX, some TREC tracks, and NTCIR have adopted an ordinal notion of relevance with documents divided into 3 or 4 classes, distinguishing slightly relevant documents from highly relevant documents. See Section 10.4 (page 210) for a detailed discussion of how this is implemented in the INEX evaluations.

One clear problem with the relevance-based assessment that we have presented is the distinction between relevance and *marginal relevance*: whether a document still has distinctive usefulness after the user has looked at certain other documents (Carbonell and Goldstein 1998). Even if a document is highly relevant, its information can be completely redundant with other documents which have already been examined. The most extreme case of this is documents that are duplicates – a phenomenon that is actually very common on the World Wide Web – but it can also easily occur when several documents provide a similar precis of an event. In such circumstances, marginal relevance is clearly a better measure of utility to the user. Maximizing marginal relevance requires returning documents that exhibit diversity and novelty. One way to approach measuring this is by using distinct facts or entities as evaluation units. This perhaps more directly measures true utility to the user but doing this makes it harder to create a test collection.

MARGINAL RELEVANCE



Exercise 8.10

[**]

Below is a table showing how two human judges rated the relevance of a set of 12 documents to a particular information need (0 = nonrelevant, 1 = relevant). Let us assume that you've written an IR system that for this query returns the set of documents {4, 5, 6, 7, 8}.

docID	Judge 1	Judge 2
1	0	0
2	0	0
3	1	1
4	1	1
5	1	0
6	1	0
7	1	0
8	1	0
9	0	1
10	0	1
11	0	1
12	0	1

- a. Calculate the kappa measure between the two judges.
- b. Calculate precision, recall, and F_1 of your system if a document is considered relevant only if the two judges agree.
- c. Calculate precision, recall, and F_1 of your system if a document is considered relevant if either judge thinks it is relevant.

8.6 A broader perspective: System quality and user utility

Formal evaluation measures are at some distance from our ultimate interest in measures of human utility: how satisfied is each user with the results the system gives for each information need that they pose? The standard way to measure human satisfaction is by various kinds of user studies. These might include quantitative measures, both objective, such as time to complete a task, as well as subjective, such as a score for satisfaction with the search engine, and qualitative measures, such as user comments on the search interface. In this section we will touch on other system aspects that allow quantitative evaluation and the issue of user utility.

8.6.1 System issues

There are many practical benchmarks on which to rate an information retrieval system beyond its retrieval quality. These include:

- How fast does it index, that is, how many documents per hour does it index for a certain distribution over document lengths? (cf. Chapter 4)
- How fast does it search, that is, what is its latency as a function of index size?
- How expressive is its query language? How fast is it on complex queries?

- How large is its document collection, in terms of the number of documents or the collection having information distributed across a broad range of topics?

All these criteria apart from query language expressiveness are straightforwardly *measurable*: we can quantify the speed or size. Various kinds of feature checklists can make query language expressiveness semi-precise.

8.6.2 User utility

What we would really like is a way of quantifying aggregate user happiness, based on the relevance, speed, and user interface of a system. One part of this is understanding the distribution of people we wish to make happy, and this depends entirely on the setting. For a web search engine, happy search users are those who find what they want. One indirect measure of such users is that they tend to return to the same engine. Measuring the rate of return of users is thus an effective metric, which would of course be more effective if you could also measure how much these users used other search engines. But advertisers are also users of modern web search engines. They are happy if customers click through to their sites and then make purchases. On an eCommerce web site, a user is likely to be wanting to purchase something. Thus, we can measure the time to purchase, or the fraction of searchers who become buyers. On a storefront web site, perhaps both the user's and the store owner's needs are satisfied if a purchase is made. Nevertheless, in general, we need to decide whether it is the end user's or the eCommerce site owner's happiness that we are trying to optimize. Usually, it is the store owner who is paying us.

For an "enterprise" (company, government, or academic) intranet search engine, the relevant metric is more likely to be user productivity: how much time do users spend looking for information that they need. There are also many other practical criteria concerning such matters as information security, which we mentioned in Section 4.6 (page 80).

User happiness is elusive to measure, and this is part of why the standard methodology uses the proxy of relevance of search results. The standard direct way to get at user satisfaction is to run user studies, where people engage in tasks, and usually various metrics are measured, the participants are observed, and ethnographic interview techniques are used to get qualitative information on satisfaction. User studies are very useful in system design, but they are time consuming and expensive to do. They are also difficult to do well, and expertise is required to design the studies and to interpret the results. We will not discuss the details of human usability testing here.

8.6.3 Refining a deployed system

If an IR system has been built and is being used by a large number of users, the system's builders can evaluate possible changes by deploying variant versions of the system and recording measures that are indicative of user satisfaction with one variant vs. others as they are being used. This method is frequently used by web search engines.

A/B TEST

The most common version of this is *A/B testing*, a term borrowed from the advertising industry. For such a test, precisely one thing is changed between the current system and a proposed system, and a small proportion of traffic (say, 1–10% of users) is randomly directed to the variant system, while most users use the current system. For example, if we wish to investigate a change to the ranking algorithm, we redirect a random sample of users to a variant system and evaluate measures such as the frequency with which people click on the top result, or any result on the first page. (This particular analysis method is referred to as *clickthrough log analysis* or *clickstream mining*. It is further discussed as a method of implicit feedback in Section 9.1.7 (page 187).)

CLICKTHROUGH LOG
ANALYSIS
CLICKSTREAM MINING

The basis of A/B testing is running a bunch of single variable tests (either in sequence or in parallel): for each test only one parameter is varied from the control (the current live system). It is therefore easy to see whether varying each parameter has a positive or negative effect. Such testing of a live system can easily and cheaply gauge the effect of a change on users, and, with a large enough user base, it is practical to measure even very small positive and negative effects. In principle, more analytic power can be achieved by varying multiple things at once in an uncorrelated (random) way, and doing standard multivariate statistical analysis, such as multiple linear regression. In practice, though, A/B testing is widely used, because A/B tests are easy to deploy, easy to understand, and easy to explain to management.

8.7 Results snippets

Having chosen or ranked the documents matching a query, we wish to present a results list that will be informative to the user. In many cases the user will not want to examine all the returned documents and so we want to make the results list informative enough that the user can do a final ranking of the documents for themselves based on relevance to their information need.³ The standard way of doing this is to provide a *snippet*, a short summary of the document, which is designed so as to allow the user to decide its relevance. Typically, the snippet consists of the document title and a short

SNIPPET

3. There are exceptions, in domains where recall is emphasized. For instance, in many legal disclosure cases, a legal associate will review *every* document that matches a keyword search.

STATIC SUMMARY
DYNAMIC SUMMARY

summary, which is automatically extracted. The question is how to design the summary so as to maximize its usefulness to the user.

The two basic kinds of summaries are *static*, which are always the same regardless of the query, and *dynamic* (or query-dependent), which are customized according to the user's information need as deduced from a query. Dynamic summaries attempt to explain why a particular document was retrieved for the query at hand.

A static summary is generally comprised of either or both a subset of the document and metadata associated with the document. The simplest form of summary takes the first two sentences or 50 words of a document, or extracts particular zones of a document, such as the title and author. Instead of zones of a document, the summary can instead use metadata associated with the document. This may be an alternative way to provide an author or date, or may include elements which are designed to give a summary, such as the *description* metadata which can appear in the *meta* element of a web HTML page. This summary is typically extracted and cached at indexing time, in such a way that it can be retrieved and presented quickly when displaying search results, whereas having to access the actual document content might be a relatively expensive operation.

TEXT SUMMARIZATION

There has been extensive work within natural language processing (NLP) on better ways to do *text summarization*. Most such work still aims only to choose sentences from the original document to present and concentrates on how to select good sentences. The models typically combine positional factors, favoring the first and last paragraphs of documents and the first and last sentences of paragraphs, with content factors, emphasizing sentences with key terms, which have low document frequency in the collection as a whole, but high frequency and good distribution across the particular document being returned. In sophisticated NLP approaches, the system synthesizes sentences for a summary, either by doing full text generation or by editing and perhaps combining sentences used in the document. For example, it might delete a relative clause or replace a pronoun with the noun phrase that it refers to. This last class of methods remains in the realm of research and is seldom used for search results: it is easier, safer, and often even better to just use sentences from the original document.

KEYWORD-IN-CONTEXT

Dynamic summaries display one or more "windows" on the document, aiming to present the pieces that have the most utility to the user in evaluating the document with respect to their information need. Usually these windows contain one or several of the query terms, and so are often referred to as *keyword-in-context* (KWIC) snippets, though sometimes they may still be pieces of the text such as the title that are selected for their query-independent information value just as in the case of static summarization. Dynamic summaries are generated in conjunction with scoring. If the query is found as a phrase, occurrences of the phrase in the document will be

... In recent years, Papua New Guinea has faced severe economic difficulties and economic growth has slowed, partly as a result of weak governance and civil war, and partly as a result of external factors such as the Bougainville civil war which led to the closure in 1989 of the Panguna mine (at that time the most important foreign exchange earner and contributor to Government finances), the Asian financial crisis, a decline in the prices of gold and copper, and a fall in the production of oil. *PNG's economic development record over the past few years is evidence that* governance issues underly many of the country's problems. Good governance, which may be defined as the transparent and accountable management of human, natural, economic and financial resources for the purposes of equitable and sustainable development, flows from proper public sector management, efficient fiscal and accounting mechanisms, and a willingness to make service delivery a priority in practice. ...

► **Figure 8.5** An example of selecting text for a dynamic snippet. This snippet was generated for a document in response to the query new guinea economic development. The figure shows in bold italic where the selected snippet text occurred in the original document.

shown as the summary. If not, windows within the document that contain multiple query terms will be selected. Commonly these windows may just stretch some number of words to the left and right of the query terms. This is a place where NLP techniques can usefully be employed: users prefer snippets that read well because they contain complete phrases.

Dynamic summaries are generally regarded as greatly improving the usability of IR systems, but they present a complication for IR system design. A dynamic summary cannot be precomputed, but, on the other hand, if a system has only a positional index, then it cannot easily reconstruct the context surrounding search engine hits in order to generate such a dynamic summary. This is one reason for using static summaries. The standard solution to this in a world of large and cheap disk drives is to locally cache all the documents at index time (notwithstanding that this approach raises various legal, information security and control issues that are far from resolved) as shown in Figure 7.5 (page 147). Then, a system can simply scan a document which is about to appear in a displayed results list to find snippets containing the query words. Beyond simply access to the text, producing a good KWIC snippet requires some care. Given a variety of keyword occurrences in a document, the goal is to choose fragments which are: (i) maximally informative about the discussion of those terms in the document, (ii) self-contained enough to be easy to read, and (iii) short enough to fit within the normally strict constraints on the space available for summaries.

Generating snippets must be fast since the system is typically generating many snippets for each query that it handles. Rather than caching an entire document, it is common to cache only a generous but fixed size prefix of the document, such as perhaps 10,000 characters. For most common, short documents, the entire document is thus cached, but huge amounts of local storage will not be wasted on potentially vast documents. Summaries of documents whose length exceeds the prefix size will be based on material in the prefix only, which is in general a useful zone in which to look for a document summary anyway.

If a document has been updated since it was last processed by a crawler and indexer, these changes will be neither in the cache nor in the index. In these circumstances, neither the index nor the summary will accurately reflect the current contents of the document, but it is the differences between the summary and the actual document content that will be more glaringly obvious to the end user.

8.8 References and further reading

Definition and implementation of the notion of relevance to a query got off to a rocky start in 1953. Swanson (1988) reports that in an evaluation in that year between two teams, they agreed that 1390 documents were variously relevant to a set of 98 questions, but disagreed on a further 1577 documents, and the disagreements were never resolved.

Rigorous formal testing of IR systems was first completed in the Cranfield experiments, beginning in the late 1950s. A retrospective discussion of the Cranfield test collection and experimentation with it can be found in (Cleverdon 1991). The other seminal series of early IR experiments were those on the SMART system by Gerard Salton and colleagues (Salton 1971b; 1991). The TREC evaluations are described in detail by Voorhees and Harman (2005). Online information is available at <http://trec.nist.gov/>. Initially, few researchers computed the statistical significance of their experimental results, but the IR community increasingly demands this (Hull 1993). User studies of IR system effectiveness began more recently (Saracevic and Kantor 1988; 1996).

F MEASURE

The notions of recall and precision were first used by Kent et al. (1955), although the term *precision* did not appear until later. The F measure (or, rather its complement $E = 1 - F$) was introduced by van Rijsbergen (1979). He provides an extensive theoretical discussion, which shows how adopting a principle of decreasing marginal relevance (at some point a user will be unwilling to sacrifice a unit of precision for an added unit of recall) leads to the harmonic mean being the appropriate method for combining precision and recall (and hence to its adoption rather than the minimum or geometric mean).

R-PRECISION

Buckley and Voorhees (2000) compare several evaluation measures, including precision at k , MAP, and R-precision, and evaluate the error rate of each measure. R-precision was adopted as the official evaluation metric in the TREC HARD track (Allan 2005). Aslam and Yilmaz (2005) examine its surprisingly close correlation to MAP, which had been noted in earlier studies (Tague-Sutcliffe and Blustein 1995, Buckley and Voorhees 2000). A standard program for evaluating IR systems which computes many measures of ranked retrieval effectiveness is Chris Buckley's `trec_eval` program used in the TREC evaluations. It can be downloaded from: http://trec.nist.gov/trec_eval/.

Kekäläinen and Järvelin (2002) argue for the superiority of graded relevance judgments when dealing with very large document collections, and Järvelin and Kekäläinen (2002) introduce cumulated gain-based methods for IR system evaluation in this context. Sakai (2007) does a study of the stability and sensitivity of evaluation measures based on graded relevance judgments from NTCIR tasks, and concludes that NDCG is best for evaluating document ranking.

Schamber et al. (1990) examine the concept of relevance, stressing its multi-dimensional and context-specific nature, but also arguing that it can be measured effectively. (Voorhees 2000) is the standard article for examining variation in relevance judgments and their effects on retrieval system scores and ranking for the TREC Ad Hoc task. Voorhees concludes that although the numbers change, the rankings are quite stable. Hersh et al. (1994) present similar analysis for a medical IR collection. In contrast, Kekäläinen (2005) analyze some of the later TRECs, exploring a 4-way relevance judgment and the notion of cumulative gain, arguing that the relevance measure used does substantially affect system rankings. See also Harter (1998). Zobel (1998) studies whether the pooling method used by TREC to collect a subset of documents that will be evaluated for relevance is reliable and fair, and concludes that it is.

KAPPA STATISTIC

The kappa statistic and its use for language-related purposes is discussed by Carletta (1996). Many standard sources (e.g., Siegel and Castellan 1988) present pooled calculation of the expected agreement, but Di Eugenio and Glass (2004) argue for preferring the unpooled agreement (though perhaps presenting multiple measures). For further discussion of alternative measures of agreement, which may in fact be better, see Lombard et al. (2002) and Krippendorff (2003).

Text summarization has been actively explored for many years. Modern work on sentence selection was initiated by Kupiec et al. (1995). More recent work includes (Barzilay and Elhadad 1997) and (Jing 2000), together with a broad selection of work appearing at the yearly DUC conferences and at other NLP venues. Tombros and Sanderson (1998) demonstrate the advantages of dynamic summaries in the IR context. Turpin et al. (2007) address how to generate snippets efficiently.

Clickthrough log analysis is studied in (Joachims 2002b, Joachims et al. 2005).

In a series of papers, Hersh, Turpin and colleagues show how improvements in formal retrieval effectiveness, as evaluated in batch experiments, do not always translate into an improved system for users (Hersh et al. 2000a;b; 2001, Turpin and Hersh 2001; 2002).

User interfaces for IR and human factors such as models of human information seeking and usability testing are outside the scope of what we cover in this book. More information on these topics can be found in other textbooks, including (Baeza-Yates and Ribeiro-Neto 1999, ch. 10) and (Korfhage 1997), and collections focused on cognitive aspects (Spink and Cole 2005).

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9 *Relevance feedback and query expansion*

SYNONYMY

In most collections, the same concept may be referred to using different words. This issue, known as *synonymy*, has an impact on the recall of most information retrieval systems. For example, you would want a search for aircraft to match plane (but only for references to an *airplane*, not a woodworking plane), and for a search on thermodynamics to match references to heat in appropriate discussions. Users often attempt to address this problem themselves by manually refining a query, as was discussed in Section 1.4; in this chapter we discuss ways in which a system can help with query refinement, either fully automatically or with the user in the loop.

The methods for tackling this problem split into two major classes: global methods and local methods. Global methods are techniques for expanding or reformulating query terms independent of the query and results returned from it, so that changes in the query wording will cause the new query to match other semantically similar terms. Global methods include:

- Query expansion/reformulation with a thesaurus or WordNet (Section 9.2.2)
- Query expansion via automatic thesaurus generation (Section 9.2.3)
- Techniques like spelling correction (discussed in Chapter 3)

Local methods adjust a query relative to the documents that initially appear to match the query. The basic methods here are:

- Relevance feedback (Section 9.1)
- Pseudo relevance feedback, also known as Blind relevance feedback (Section 9.1.6)
- (Global) indirect relevance feedback (Section 9.1.7)

In this chapter, we will mention all of these approaches, but we will concentrate on relevance feedback, which is one of the most used and most successful approaches.

9.1 Relevance feedback and pseudo relevance feedback

RELEVANCE FEEDBACK

The idea of *relevance feedback* (RF) is to involve the user in the retrieval process so as to improve the final result set. In particular, the user gives feedback on the relevance of documents in an initial set of results. The basic procedure is:

- The user issues a (short, simple) query.
- The system returns an initial set of retrieval results.
- The user marks some returned documents as relevant or nonrelevant.
- The system computes a better representation of the information need based on the user feedback.
- The system displays a revised set of retrieval results.

Relevance feedback can go through one or more iterations of this sort. The process exploits the idea that it may be difficult to formulate a good query when you don't know the collection well, but it is easy to judge particular documents, and so it makes sense to engage in iterative query refinement of this sort. In such a scenario, relevance feedback can also be effective in tracking a user's evolving information need: seeing some documents may lead users to refine their understanding of the information they are seeking.

Image search provides a good example of relevance feedback. Not only is it easy to see the results at work, but this is a domain where a user can easily have difficulty formulating what they want in words, but can easily indicate relevant or nonrelevant images. After the user enters an initial query for bike on the demonstration system at:

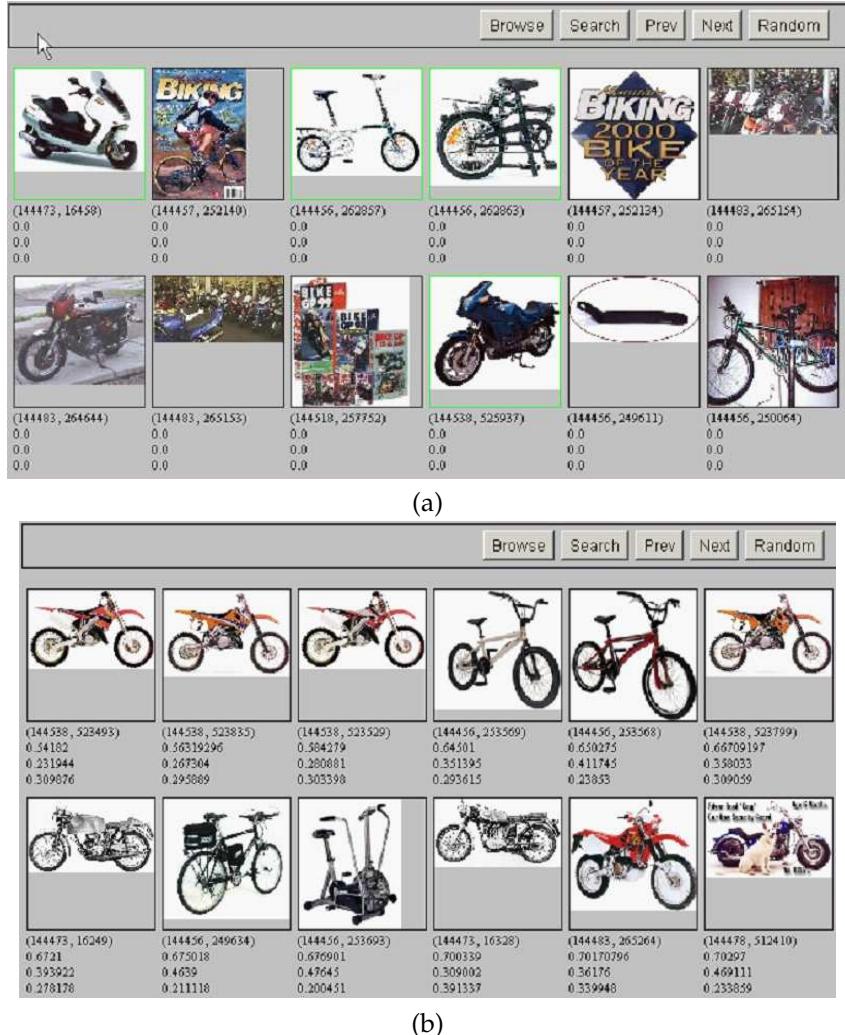
<http://nayana.ece.ucsb.edu/imsearch/imsearch.html>

the initial results (in this case, images) are returned. In Figure 9.1 (a), the user has selected some of them as relevant. These will be used to refine the query, while other displayed results have no effect on the reformulation. Figure 9.1 (b) then shows the new top-ranked results calculated after this round of relevance feedback.

Figure 9.2 shows a textual IR example where the user wishes to find out about new applications of space satellites.

9.1.1 The Rocchio algorithm for relevance feedback

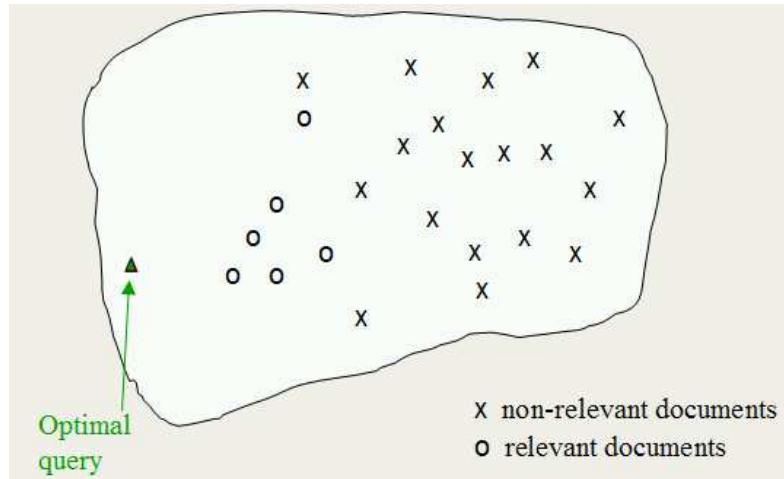
The Rocchio Algorithm is the classic algorithm for implementing relevance feedback. It models a way of incorporating relevance feedback information into the vector space model of Section 6.3.



► **Figure 9.1** Relevance feedback searching over images. (a) The user views the initial query results for a query of *bike*, selects the first, third and fourth result in the top row and the fourth result in the bottom row as relevant, and submits this feedback. (b) The users sees the revised result set. Precision is greatly improved. From <http://nayana.ece.ucsb.edu/imsearch/imsearch.html> (Newsam et al. 2001).

- (a) Query: New space satellite applications
- (b)
 - + 1. 0.539, 08/13/91, NASA Hasn't Scrapped Imaging Spectrometer
 - + 2. 0.533, 07/09/91, NASA Scratches Environment Gear From Satellite Plan
 - 3. 0.528, 04/04/90, Science Panel Backs NASA Satellite Plan, But Urges Launches of Smaller Probes
 - 4. 0.526, 09/09/91, A NASA Satellite Project Accomplishes Incredible Feat: Staying Within Budget
 - 5. 0.525, 07/24/90, Scientist Who Exposed Global Warming Proposes Satellites for Climate Research
 - 6. 0.524, 08/22/90, Report Provides Support for the Critics Of Using Big Satellites to Study Climate
 - 7. 0.516, 04/13/87, Arianespace Receives Satellite Launch Pact From Telesat Canada
 - + 8. 0.509, 12/02/87, Telecommunications Tale of Two Companies
- (c)
 - 2.074 new 15.106 space
 - 30.816 satellite 5.660 application
 - 5.991 nasa 5.196 eos
 - 4.196 launch 3.972 aster
 - 3.516 instrument 3.446 arianespace
 - 3.004 bundespost 2.806 ss
 - 2.790 rocket 2.053 scientist
 - 2.003 broadcast 1.172 earth
 - 0.836 oil 0.646 measure
- (d)
 - * 1. 0.513, 07/09/91, NASA Scratches Environment Gear From Satellite Plan
 - * 2. 0.500, 08/13/91, NASA Hasn't Scrapped Imaging Spectrometer
 - 3. 0.493, 08/07/89, When the Pentagon Launches a Secret Satellite, Space Sleuths Do Some Spy Work of Their Own
 - 4. 0.493, 07/31/89, NASA Uses 'Warm' Superconductors For Fast Circuit
 - * 5. 0.492, 12/02/87, Telecommunications Tale of Two Companies
 - 6. 0.491, 07/09/91, Soviets May Adapt Parts of SS-20 Missile For Commercial Use
 - 7. 0.490, 07/12/88, Gaping Gap: Pentagon Lags in Race To Match the Soviets In Rocket Launchers
 - 8. 0.490, 06/14/90, Rescue of Satellite By Space Agency To Cost \$90 Million

► **Figure 9.2** Example of relevance feedback on a text collection. (a) The initial query (a). (b) The user marks some relevant documents (shown with a plus sign). (c) The query is then expanded by 18 terms with weights as shown. (d) The revised top results are then shown. A * marks the documents which were judged relevant in the relevance feedback phase.



► **Figure 9.3** The Rocchio optimal query for separating relevant and nonrelevant documents.

The underlying theory. We want to find a query vector, denoted as \vec{q} , that maximizes similarity with relevant documents while minimizing similarity with nonrelevant documents. If C_r is the set of relevant documents and C_{nr} is the set of nonrelevant documents, then we wish to find:¹

$$(9.1) \quad \vec{q}_{opt} = \arg \max_{\vec{q}} [\text{sim}(\vec{q}, C_r) - \text{sim}(\vec{q}, C_{nr})],$$

where sim is defined as in Equation 6.10. Under cosine similarity, the optimal query vector \vec{q}_{opt} for separating the relevant and nonrelevant documents is:

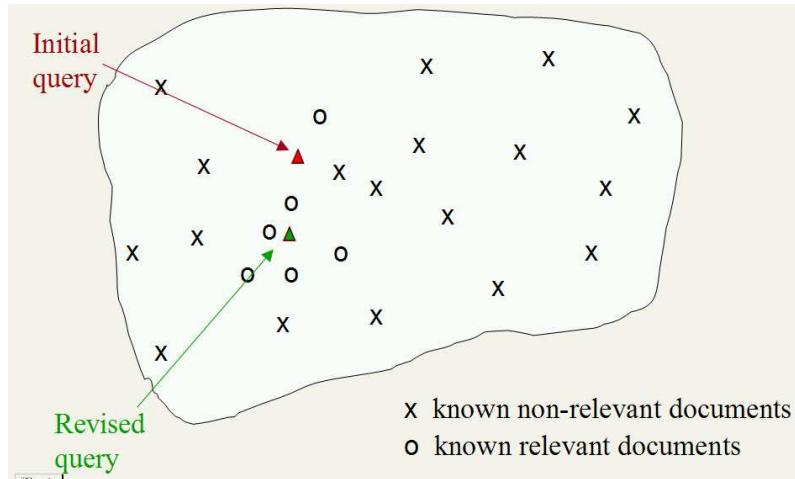
$$(9.2) \quad \vec{q}_{opt} = \frac{1}{|C_r|} \sum_{\vec{d}_j \in C_r} \vec{d}_j - \frac{1}{|C_{nr}|} \sum_{\vec{d}_j \in C_{nr}} \vec{d}_j$$

That is, the optimal query is the vector difference between the centroids of the relevant and nonrelevant documents; see Figure 9.3. However, this observation is not terribly useful, precisely because the full set of relevant documents is not known: it is what we want to find.

ROCCIO ALGORITHM

The Rocchio (1971) algorithm. This was the relevance feedback mecha-

1. In the equation, $\arg \max_x f(x)$ returns a value of x which maximizes the value of the function $f(x)$. Similarly, $\arg \min_x f(x)$ returns a value of x which minimizes the value of the function $f(x)$.



► **Figure 9.4** An application of Rocchio’s algorithm. Some documents have been labeled as relevant and nonrelevant and the initial query vector is moved in response to this feedback.

nism introduced in and popularized by Salton’s SMART system around 1970. In a real IR query context, we have a user query and partial knowledge of known relevant and nonrelevant documents. The algorithm proposes using the modified query \vec{q}_m :

$$(9.3) \quad \vec{q}_m = \alpha \vec{q}_0 + \beta \frac{1}{|D_r|} \sum_{\vec{d}_j \in D_r} \vec{d}_j - \gamma \frac{1}{|D_{nr}|} \sum_{\vec{d}_j \in D_{nr}} \vec{d}_j$$

where q_0 is the original query vector, D_r and D_{nr} are the set of known relevant and nonrelevant documents respectively, and α , β , and γ are weights attached to each term. These control the balance between trusting the judged document set versus the query: if we have a lot of judged documents, we would like a higher β and γ . Starting from q_0 , the new query moves you some distance toward the centroid of the relevant documents and some distance away from the centroid of the nonrelevant documents. This new query can be used for retrieval in the standard vector space model (see Section 6.3). We can easily leave the positive quadrant of the vector space by subtracting off a nonrelevant document’s vector. In the Rocchio algorithm, negative term weights are ignored. That is, the term weight is set to 0. Figure 9.4 shows the effect of applying relevance feedback.

Relevance feedback can improve both recall and precision. But, in practice, it has been shown to be most useful for increasing recall in situations

where recall is important. This is partly because the technique expands the query, but it is also partly an effect of the use case: when they want high recall, users can be expected to take time to review results and to iterate on the search. Positive feedback also turns out to be much more valuable than negative feedback, and so most IR systems set $\gamma < \beta$. Reasonable values might be $\alpha = 1$, $\beta = 0.75$, and $\gamma = 0.15$. In fact, many systems, such as the image search system in Figure 9.1, allow only positive feedback, which is equivalent to setting $\gamma = 0$. Another alternative is to use only the marked nonrelevant document which received the highest ranking from the IR system as negative feedback (here, $|D_{nr}| = 1$ in Equation (9.3)). While many of the experimental results comparing various relevance feedback variants are rather inconclusive, some studies have suggested that this variant, called *Ide dec-hi* is the most effective or at least the most consistent performer.



9.1.2 Probabilistic relevance feedback

Rather than reweighting the query in a vector space, if a user has told us some relevant and nonrelevant documents, then we can proceed to build a classifier. One way of doing this is with a Naive Bayes probabilistic model. If R is a Boolean indicator variable expressing the relevance of a document, then we can estimate $P(x_t = 1|R)$, the probability of a term t appearing in a document, depending on whether it is relevant or not, as:

$$(9.4) \quad \begin{aligned} \hat{P}(x_t = 1|R = 1) &= |VR_t|/|VR| \\ \hat{P}(x_t = 1|R = 0) &= (df_t - |VR_t|)/(N - |VR|) \end{aligned}$$

where N is the total number of documents, df_t is the number that contain t , VR is the set of known relevant documents, and VR_t is the subset of this set containing t . Even though the set of known relevant documents is a perhaps small subset of the true set of relevant documents, if we assume that the set of relevant documents is a small subset of the set of all documents then the estimates given above will be reasonable. This gives a basis for another way of changing the query term weights. We will discuss such probabilistic approaches more in Chapters 11 and 13, and in particular outline the application to relevance feedback in Section 11.3.4 (page 228). For the moment, observe that using just Equation (9.4) as a basis for term-weighting is likely insufficient. The equations use only collection statistics and information about the term distribution within the documents judged relevant. They preserve no memory of the original query.

9.1.3 When does relevance feedback work?

The success of relevance feedback depends on certain assumptions. Firstly, the user has to have sufficient knowledge to be able to make an initial query

which is at least somewhere close to the documents they desire. This is needed anyhow for successful information retrieval in the basic case, but it is important to see the kinds of problems that relevance feedback cannot solve alone. Cases where relevance feedback alone is not sufficient include:

- Misspellings. If the user spells a term in a different way to the way it is spelled in any document in the collection, then relevance feedback is unlikely to be effective. This can be addressed by the spelling correction techniques of Chapter 3.
- Cross-language information retrieval. Documents in another language are not nearby in a vector space based on term distribution. Rather, documents in the same language cluster more closely together.
- Mismatch of searcher’s vocabulary versus collection vocabulary. If the user searches for laptop but all the documents use the term notebook computer, then the query will fail, and relevance feedback is again most likely ineffective.

Secondly, the relevance feedback approach requires relevant documents to be similar to each other. That is, they should cluster. Ideally, the term distribution in all relevant documents will be similar to that in the documents marked by the users, while the term distribution in all nonrelevant documents will be different from those in relevant documents. Things will work well if all relevant documents are tightly clustered around a single prototype, or, at least, if there are different prototypes, if the relevant documents have significant vocabulary overlap, while similarities between relevant and nonrelevant documents are small. Implicitly, the Rocchio relevance feedback model treats relevant documents as a single *cluster*, which it models via the centroid of the cluster. This approach does not work as well if the relevant documents are a multimodal class, that is, they consist of several clusters of documents within the vector space. This can happen with:

- Subsets of the documents using different vocabulary, such as Burma vs. Myanmar
- A query for which the answer set is inherently disjunctive, such as Pop stars who once worked at Burger King.
- Instances of a general concept, which often appear as a disjunction of more specific concepts, for example, felines.

Good editorial content in the collection can often provide a solution to this problem. For example, an article on the attitudes of different groups to the situation in Burma could introduce the terminology used by different parties, thus linking the document clusters.

Relevance feedback is not necessarily popular with users. Users are often reluctant to provide explicit feedback, or in general do not wish to prolong the search interaction. Furthermore, it is often harder to understand why a particular document was retrieved after relevance feedback is applied.

Relevance feedback can also have practical problems. The long queries that are generated by straightforward application of relevance feedback techniques are inefficient for a typical IR system. This results in a high computing cost for the retrieval and potentially long response times for the user. A partial solution to this is to only reweight certain prominent terms in the relevant documents, such as perhaps the top 20 terms by term frequency. Some experimental results have also suggested that using a limited number of terms like this may give better results (Harman 1992) though other work has suggested that using more terms is better in terms of retrieved document quality (Buckley et al. 1994b).

9.1.4 Relevance feedback on the web

Some web search engines offer a similar/related pages feature: the user indicates a document in the results set as exemplary from the standpoint of meeting his information need and requests more documents like it. This can be viewed as a particular simple form of relevance feedback. However, in general relevance feedback has been little used in web search. One exception was the Excite web search engine, which initially provided full relevance feedback. However, the feature was in time dropped, due to lack of use. On the web, few people use advanced search interfaces and most would like to complete their search in a single interaction. But the lack of uptake also probably reflects two other factors: relevance feedback is hard to explain to the average user, and relevance feedback is mainly a recall enhancing strategy, and web search users are only rarely concerned with getting sufficient recall.

Spink et al. (2000) present results from the use of relevance feedback in the Excite search engine. Only about 4% of user query sessions used the relevance feedback option, and these were usually exploiting the “More like this” link next to each result. About 70% of users only looked at the first page of results and did not pursue things any further. For people who used relevance feedback, results were improved about two thirds of the time.

An important more recent thread of work is the use of clickstream data (what links a user clicks on) to provide indirect relevance feedback. Use of this data is studied in detail in (Joachims 2002b, Joachims et al. 2005). The very successful use of web link structure (see Chapter 21) can also be viewed as implicit feedback, but provided by page authors rather than readers (though in practice most authors are also readers).

**Exercise 9.1**

In Rocchio's algorithm, what weight setting for $\alpha/\beta/\gamma$ does a "Find pages like this one" search correspond to?

Exercise 9.2

[*]

Give three reasons why relevance feedback has been little used in web search.

9.1.5 Evaluation of relevance feedback strategies

Interactive relevance feedback can give very substantial gains in retrieval performance. Empirically, one round of relevance feedback is often very useful. Two rounds is sometimes marginally more useful. Successful use of relevance feedback requires enough judged documents, otherwise the process is unstable in that it may drift away from the user's information need. Accordingly, having at least five judged documents is recommended.

There is some subtlety to evaluating the effectiveness of relevance feedback in a sound and enlightening way. The obvious first strategy is to start with an initial query q_0 and to compute a precision-recall graph. Following one round of feedback from the user, we compute the modified query q_m and again compute a precision-recall graph. Here, in both rounds we assess performance over all documents in the collection, which makes comparisons straightforward. If we do this, we find spectacular gains from relevance feedback: gains on the order of 50% in mean average precision. But unfortunately it is cheating. The gains are partly due to the fact that known relevant documents (judged by the user) are now ranked higher. Fairness demands that we should only evaluate with respect to documents not seen by the user.

A second idea is to use documents in the *residual collection* (the set of documents minus those assessed relevant) for the second round of evaluation. This seems like a more realistic evaluation. Unfortunately, the measured performance can then often be lower than for the original query. This is particularly the case if there are few relevant documents, and so a fair proportion of them have been judged by the user in the first round. The relative performance of variant relevance feedback methods can be validly compared, but it is difficult to validly compare performance with and without relevance feedback because the collection size and the number of relevant documents changes from before the feedback to after it.

Thus neither of these methods is fully satisfactory. A third method is to have two collections, one which is used for the initial query and relevance judgments, and the second that is then used for comparative evaluation. The performance of both q_0 and q_m can be validly compared on the second collection.

Perhaps the best evaluation of the utility of relevance feedback is to do user studies of its effectiveness, in particular by doing a time-based comparison:

Term weighting	Precision at $k = 50$	
	no RF	pseudo RF
lnc.ltc	64.2%	72.7%
Lnu.ltu	74.2%	87.0%

► **Figure 9.5** Results showing pseudo relevance feedback greatly improving performance. These results are taken from the Cornell SMART system at TREC 4 (Buckley et al. 1995), and also contrast the use of two different length normalization schemes (L vs. l); cf. Figure 6.15 (page 128). Pseudo relevance feedback consisted of adding 20 terms to each query.

how fast does a user find relevant documents with relevance feedback vs. another strategy (such as query reformulation), or alternatively, how many relevant documents does a user find in a certain amount of time. Such notions of user utility are fairest and closest to real system usage.

9.1.6 Pseudo relevance feedback

PSEUDO RELEVANCE
FEEDBACK
BLIND RELEVANCE
FEEDBACK

Pseudo relevance feedback, also known as *blind relevance feedback*, provides a method for automatic local analysis. It automates the manual part of relevance feedback, so that the user gets improved retrieval performance without an extended interaction. The method is to do normal retrieval to find an initial set of most relevant documents, to then *assume* that the top k ranked documents are relevant, and finally to do relevance feedback as before under this assumption.

This automatic technique mostly works. Evidence suggests that it tends to work better than global analysis (Section 9.2). It has been found to improve performance in the TREC ad hoc task. See for example the results in Figure 9.5. But it is not without the dangers of an automatic process. For example, if the query is about copper mines and the top several documents are all about mines in Chile, then there may be query drift in the direction of documents on Chile.

9.1.7 Indirect relevance feedback

IMPLICIT RELEVANCE
FEEDBACK

We can also use indirect sources of evidence rather than explicit feedback on relevance as the basis for relevance feedback. This is often called *implicit (relevance) feedback*. Implicit feedback is less reliable than explicit feedback, but is more useful than pseudo relevance feedback, which contains no evidence of user judgments. Moreover, while users are often reluctant to provide explicit feedback, it is easy to collect implicit feedback in large quantities for a high volume system, such as a web search engine.

On the web, DirectHit introduced the idea of ranking more highly documents that users chose to look at more often. In other words, clicks on links were assumed to indicate that the page was likely relevant to the query. This approach makes various assumptions, such as that the document summaries displayed in results lists (on whose basis users choose which documents to click on) are indicative of the relevance of these documents. In the original DirectHit search engine, the data about the click rates on pages was gathered globally, rather than being user or query specific. This is one form of the general area of *clickstream mining*. Today, a closely related approach is used in ranking the advertisements that match a web search query (Chapter 19).

9.1.8 Summary

Relevance feedback has been shown to be very effective at improving relevance of results. Its successful use requires queries for which the set of relevant documents is medium to large. Full relevance feedback is often onerous for the user, and its implementation is not very efficient in most IR systems. In many cases, other types of interactive retrieval may improve relevance by about as much with less work.

Beyond the core ad hoc retrieval scenario, other uses of relevance feedback include:

- Following a changing information need (e.g., names of car models of interest change over time)
- Maintaining an information filter (e.g., for a news feed). Such filters are discussed further in Chapter 13.
- Active learning (deciding which examples it is most useful to know the class of to reduce annotation costs).



Exercise 9.3

Under what conditions would the modified query q_m in Equation 9.3 be the same as the original query q_0 ? In all other cases, is q_m closer than q_0 to the centroid of the relevant documents?

Exercise 9.4

Why is positive feedback likely to be more useful than negative feedback to an IR system? Why might only using one nonrelevant document be more effective than using several?

Exercise 9.5

Suppose that a user's initial query is cheap CDs cheap DVDs extremely cheap CDs. The user examines two documents, d_1 and d_2 . She judges d_1 , with the content *CDs cheap software cheap CDs* relevant and d_2 with content *cheap thrills DVDs* nonrelevant. Assume that we are using direct term frequency (with no scaling and no document

frequency). There is no need to length-normalize vectors. Using Rocchio relevance feedback as in Equation (9.3) what would the revised query vector be after relevance feedback? Assume $\alpha = 1, \beta = 0.75, \gamma = 0.25$.

Exercise 9.6[\star]

Omar has implemented a relevance feedback web search system, where he is going to do relevance feedback based only on words in the title text returned for a page (for efficiency). The user is going to rank 3 results. The first user, Jinxing, queries for:

banana slug

and the top three titles returned are:

banana slug Ariolimax columbianus
Santa Cruz mountains banana slug
Santa Cruz Campus Mascot

Jinxing judges the first two documents relevant, and the third nonrelevant. Assume that Omar's search engine uses term frequency but no length normalization nor IDF. Assume that he is using the Rocchio relevance feedback mechanism, with $\alpha = \beta = \gamma = 1$. Show the final revised query that would be run. (Please list the vector elements in alphabetical order.)

9.2 Global methods for query reformulation

In this section we more briefly discuss three global methods for expanding a query: by simply aiding the user in doing so, by using a manual thesaurus, and through building a thesaurus automatically.

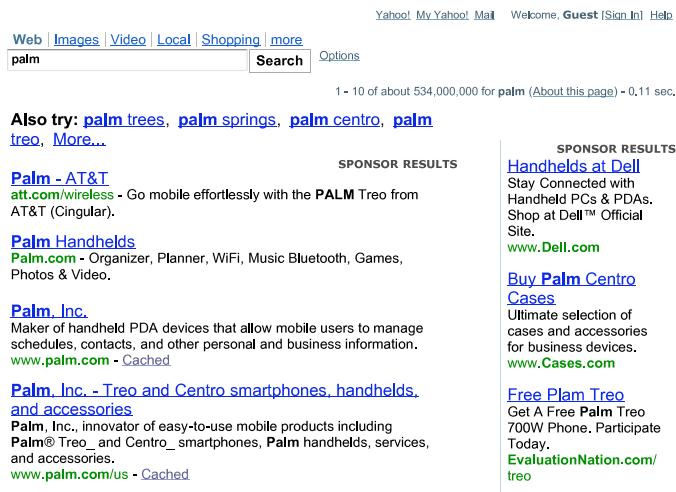
9.2.1 Vocabulary tools for query reformulation

Various user supports in the search process can help the user see how their searches are or are not working. This includes information about words that were omitted from the query because they were on stop lists, what words were stemmed to, the number of hits on each term or phrase, and whether words were dynamically turned into phrases. The IR system might also suggest search terms by means of a thesaurus or a controlled vocabulary. A user can also be allowed to browse lists of the terms that are in the inverted index, and thus find good terms that appear in the collection.

9.2.2 Query expansion

QUERY EXPANSION

In relevance feedback, users give additional input on documents (by marking documents in the results set as relevant or not), and this input is used to reweight the terms in the query for documents. In *query expansion* on the other hand, users give additional input on query words or phrases, possibly suggesting additional query terms. Some search engines (especially on the



► **Figure 9.6** An example of query expansion in the interface of the Yahoo! web search engine in 2006. The expanded query suggestions appear just below the “Search Results” bar.

web) suggest related queries in response to a query; the users then opt to use one of these alternative query suggestions. Figure 9.6 shows an example of query suggestion options being presented in the Yahoo! web search engine. The central question in this form of query expansion is how to generate alternative or expanded queries for the user. The most common form of query expansion is global analysis, using some form of thesaurus. For each term t in a query, the query can be automatically expanded with synonyms and related words of t from the thesaurus. Use of a thesaurus can be combined with ideas of term weighting: for instance, one might weight added terms less than original query terms.

Methods for building a thesaurus for query expansion include:

- Use of a controlled vocabulary that is maintained by human editors. Here, there is a canonical term for each concept. The subject headings of traditional library subject indexes, such as the Library of Congress Subject Headings, or the Dewey Decimal system are examples of a controlled vocabulary. Use of a controlled vocabulary is quite common for well-resourced domains. A well-known example is the Unified Medical Language System (UMLS) used with MedLine for querying the biomedical research literature. For example, in Figure 9.7, neoplasms was added to a

- User query: cancer
- PubMed query: ("neoplasms"[TIAB] NOT Medline[SB]) OR "neoplasms"[MeSH Terms] OR cancer[Text Word]
- User query: skin itch
- PubMed query: ("skin"[MeSH Terms] OR ("integumentary system"[TIAB] NOT Medline[SB]) OR "integumentary system"[MeSH Terms] OR skin[Text Word]) AND ((("pruritus"[TIAB] NOT Medline[SB]) OR "pruritus"[MeSH Terms] OR itch[Text Word]))

► **Figure 9.7** Examples of query expansion via the PubMed thesaurus. When a user issues a query on the PubMed interface to Medline at <http://www.ncbi.nlm.nih.gov/entrez/>, their query is mapped on to the Medline vocabulary as shown.

search for cancer. This Medline query expansion also contrasts with the Yahoo! example. The Yahoo! interface is a case of interactive query expansion, whereas PubMed does automatic query expansion. Unless the user chooses to examine the submitted query, they may not even realize that query expansion has occurred.

- A manual thesaurus. Here, human editors have built up sets of synonymous names for concepts, without designating a canonical term. The UMLS metathesaurus is one example of a thesaurus. Statistics Canada maintains a thesaurus of preferred terms, synonyms, broader terms, and narrower terms for matters on which the government collects statistics, such as goods and services. This thesaurus is also bilingual English and French.
- An automatically derived thesaurus. Here, word co-occurrence statistics over a collection of documents in a domain are used to automatically induce a thesaurus; see Section 9.2.3.
- Query reformulations based on query log mining. Here, we exploit the manual query reformulations of other users to make suggestions to a new user. This requires a huge query volume, and is thus particularly appropriate to web search.

Thesaurus-based query expansion has the advantage of not requiring any user input. Use of query expansion generally increases recall and is widely used in many science and engineering fields. As well as such global analysis techniques, it is also possible to do query expansion by local analysis, for instance, by analyzing the documents in the result set. User input is now

Word	Nearest neighbors
absolutely	absurd, whatsoever, totally, exactly, nothing
bottomed	dip, copper, drops, topped, slide, trimmed
captivating	shimmer, stunningly, superbly, plucky, witty
doghouse	dog, porch, crawling, beside, downstairs
makeup	repellent, lotion, glossy, sunscreen, skin, gel
mediating	reconciliation, negotiate, case, conciliation
keeping	hoping, bring, wiping, could, some, would
lithographs	drawings, Picasso, Dali, sculptures, Gauguin
pathogens	toxins, bacteria, organisms, bacterial, parasite
senses	grasp, psyche, truly, clumsy, naive, innate

► **Figure 9.8** An example of an automatically generated thesaurus. This example is based on the work in Schütze (1998), which employs latent semantic indexing (see Chapter 18).

usually required, but a distinction remains as to whether the user is giving feedback on documents or on query terms.

9.2.3 Automatic thesaurus generation

As an alternative to the cost of a manual thesaurus, we could attempt to generate a thesaurus automatically by analyzing a collection of documents. There are two main approaches. One is simply to exploit word cooccurrence. We say that words co-occurring in a document or paragraph are likely to be in some sense similar or related in meaning, and simply count text statistics to find the most similar words. The other approach is to use a shallow grammatical analysis of the text and to exploit grammatical relations or grammatical dependencies. For example, we say that entities that are grown, cooked, eaten, and digested, are more likely to be food items. Simply using word cooccurrence is more robust (it cannot be misled by parser errors), but using grammatical relations is more accurate.

The simplest way to compute a co-occurrence thesaurus is based on term-term similarities. We begin with a term-document matrix A , where each cell $A_{t,d}$ is a weighted count $w_{t,d}$ for term t and document d , with weighting so A has length-normalized rows. If we then calculate $C = AA^T$, then $C_{u,v}$ is a similarity score between terms u and v , with a larger number being better. Figure 9.8 shows an example of a thesaurus derived in basically this manner, but with an extra step of dimensionality reduction via Latent Semantic Indexing, which we discuss in Chapter 18. While some of the thesaurus terms are good or at least suggestive, others are marginal or bad. The quality of the associations is typically a problem. Term ambiguity easily introduces irrel-

event statistically correlated terms. For example, a query for Apple computer may expand to Apple red fruit computer. In general these thesauri suffer from both false positives and false negatives. Moreover, since the terms in the automatic thesaurus are highly correlated in documents anyway (and often the collection used to derive the thesaurus is the same as the one being indexed), this form of query expansion may not retrieve many additional documents.

Query expansion is often effective in increasing recall. However, there is a high cost to manually producing a thesaurus and then updating it for scientific and terminological developments within a field. In general a domain-specific thesaurus is required: general thesauri and dictionaries give far too little coverage of the rich domain-particular vocabularies of most scientific fields. However, query expansion may also significantly decrease precision, particularly when the query contains ambiguous terms. For example, if the user searches for interest rate, expanding the query to interest rate fascinate evaluate is unlikely to be useful. Overall, query expansion is less successful than relevance feedback, though it may be as good as pseudo relevance feedback. It does, however, have the advantage of being much more understandable to the system user.

**Exercise 9.7**

If A is simply a Boolean cooccurrence matrix, then what do you get as the entries in C ?

9.3 References and further reading

Work in information retrieval quickly confronted the problem of variant expression which meant that the words in a query might not appear in a document, despite it being relevant to the query. An early experiment about 1960 cited by Swanson (1988) found that only 11 out of 23 documents properly indexed under the subject *toxicity* had any use of a word containing the stem *toxi*. There is also the issue of translation, of users knowing what terms a document will use. Blair and Maron (1985) conclude that “it is impossibly difficult for users to predict the exact words, word combinations, and phrases that are used by all (or most) relevant documents and only (or primarily) by those documents”.

The main initial papers on relevance feedback using vector space models all appear in Salton (1971b), including the presentation of the Rocchio algorithm (Rocchio 1971) and the Ide dec-hi variant along with evaluation of several variants (Ide 1971). Another variant is to regard *all* documents in the collection apart from those judged relevant as nonrelevant, rather than only ones that are explicitly judged nonrelevant. However, Schütze et al. (1995) and Singhal et al. (1997) show that better results are obtained for routing by using only documents close to the query of interest rather than all

documents. Other later work includes Salton and Buckley (1990), Riezler et al. (2007) (a statistical NLP approach to RF) and the recent survey paper Ruthven and Lalmas (2003).

The effectiveness of interactive relevance feedback systems is discussed in (Salton 1989, Harman 1992, Buckley et al. 1994b). Koenemann and Belkin (1996) do user studies of the effectiveness of relevance feedback.

Traditionally Roget's thesaurus has been the best known English language thesaurus (Roget 1946). In recent computational work, people almost always use WordNet (Fellbaum 1998), not only because it is free, but also because of its rich link structure. It is available at: <http://wordnet.princeton.edu>.

Qiu and Frei (1993) and Schütze (1998) discuss automatic thesaurus generation. Xu and Croft (1996) explore using both local and global query expansion.

13 *Text classification and Naive Bayes*

STANDING QUERY

Thus far, this book has mainly discussed the process of *ad hoc retrieval*, where users have transient information needs that they try to address by posing one or more queries to a search engine. However, many users have ongoing information needs. For example, you might need to track developments in *multicore computer chips*. One way of doing this is to issue the query *multicore AND computer AND chip* against an index of recent newswire articles each morning. In this and the following two chapters we examine the question: How can this repetitive task be automated? To this end, many systems support *standing queries*. A standing query is like any other query except that it is periodically executed on a collection to which new documents are incrementally added over time.

If your standing query is just *multicore AND computer AND chip*, you will tend to miss many relevant new articles which use other terms such as *multicore processors*. To achieve good recall, standing queries thus have to be refined over time and can gradually become quite complex. In this example, using a Boolean search engine with stemming, you might end up with a query like (*multicore OR multi-core*) AND (*chip OR processor OR microprocessor*).

CLASSIFICATION

To capture the generality and scope of the problem space to which standing queries belong, we now introduce the general notion of a *classification* problem. Given a set of *classes*, we seek to determine which class(es) a given object belongs to. In the example, the standing query serves to divide new newswire articles into the two classes: documents about multicore computer chips and documents not about multicore computer chips. We refer to this as *two-class classification*. Classification using standing queries is also called *routing* or *filtering* and will be discussed further in Section 15.3.1 (page 335).

ROUTING
FILTERING

TEXT CLASSIFICATION

A class need not be as narrowly focused as the standing query *multicore computer chips*. Often, a class is a more general subject area like *China* or *coffee*. Such more general classes are usually referred to as *topics*, and the classification task is then called *text classification*, *text categorization*, *topic classification*, or *topic spotting*. An example for *China* appears in Figure 13.1. Standing queries and topics differ in their degree of specificity, but the methods for

solving routing, filtering, and text classification are essentially the same. We therefore include routing and filtering under the rubric of text classification in this and the following chapters.

The notion of classification is very general and has many applications within and beyond information retrieval (IR). For instance, in computer vision, a classifier may be used to divide images into classes such as *landscape*, *portrait*, and *neither*. We focus here on examples from information retrieval such as:

- Several of the preprocessing steps necessary for indexing as discussed in Chapter 2: detecting a document's encoding (ASCII, Unicode UTF-8 etc; page 20); word segmentation (Is the white space between two letters a word boundary or not? page 24) ; truecasing (page 30); and identifying the language of a document (page 46).
- The automatic detection of spam pages (which then are not included in the search engine index).
- The automatic detection of sexually explicit content (which is included in search results only if the user turns an option such as SafeSearch off).
- *Sentiment detection* or the automatic classification of a movie or product review as positive or negative. An example application is a user searching for negative reviews before buying a camera to make sure it has no undesirable features or quality problems.

SENTIMENT DETECTION

EMAIL SORTING

VERTICAL SEARCH ENGINE

- Personal *email sorting*. A user may have folders like *talk announcements*, *electronic bills*, *email from family and friends*, and so on, and may want a classifier to classify each incoming email and automatically move it to the appropriate folder. It is easier to find messages in sorted folders than in a very large inbox. The most common case of this application is a spam folder that holds all suspected spam messages.
- Topic-specific or *vertical* search. *Vertical search engines* restrict searches to a particular topic. For example, the query computer science on a vertical search engine for the topic *China* will return a list of Chinese computer science departments with higher precision and recall than the query computer science China on a general purpose search engine. This is because the vertical search engine does not include web pages in its index that contain the term *china* in a different sense (e.g., referring to a hard white ceramic), but does include relevant pages even if they do not explicitly mention the term *China*.
- Finally, the ranking function in ad hoc information retrieval can also be based on a document classifier as we will explain in Section 15.4 (page 341).

This list shows the general importance of classification in IR. Most retrieval systems today contain multiple components that use some form of classifier. The classification task we will use as an example in this book is text classification.

A computer is not essential for classification. Many classification tasks have traditionally been solved manually. Books in a library are assigned Library of Congress categories by a librarian. But manual classification is expensive to scale. The *multicore computer chips* example illustrates one alternative approach: classification by the use of standing queries – which can be thought of as *rules* – most commonly written by hand. As in our example (multicore OR multi-core) AND (chip OR processor OR microprocessor), rules are sometimes equivalent to Boolean expressions.

A rule captures a certain combination of keywords that indicates a class. Hand-coded rules have good scaling properties, but creating and maintaining them over time is labor intensive. A technically skilled person (e.g., a domain expert who is good at writing regular expressions) can create rule sets that will rival or exceed the accuracy of the automatically generated classifiers we will discuss shortly; however, it can be hard to find someone with this specialized skill.

Apart from manual classification and hand-crafted rules, there is a third approach to text classification, namely, machine learning-based text classification. It is the approach that we focus on in the next several chapters. In machine learning, the set of rules or, more generally, the decision criterion of the text classifier, is learned automatically from training data. This approach is also called *statistical text classification* if the learning method is statistical. In statistical text classification, we require a number of good example documents (or training documents) for each class. The need for manual classification is not eliminated because the training documents come from a person who has labeled them – where *labeling* refers to the process of annotating each document with its class. But labeling is arguably an easier task than writing rules. Almost anybody can look at a document and decide whether or not it is related to China. Sometimes such labeling is already implicitly part of an existing workflow. For instance, you may go through the news articles returned by a standing query each morning and give relevance feedback (cf. Chapter 9) by moving the relevant articles to a special folder like *multicore-processors*.

We begin this chapter with a general introduction to the text classification problem including a formal definition (Section 13.1); we then cover Naive Bayes, a particularly simple and effective classification method (Sections 13.2–13.4). All of the classification algorithms we study represent documents in high-dimensional spaces. To improve the efficiency of these algorithms, it is generally desirable to reduce the dimensionality of these spaces; to this end, a technique known as *feature selection* is commonly applied in text clas-

RULES IN TEXT
CLASSIFICATION

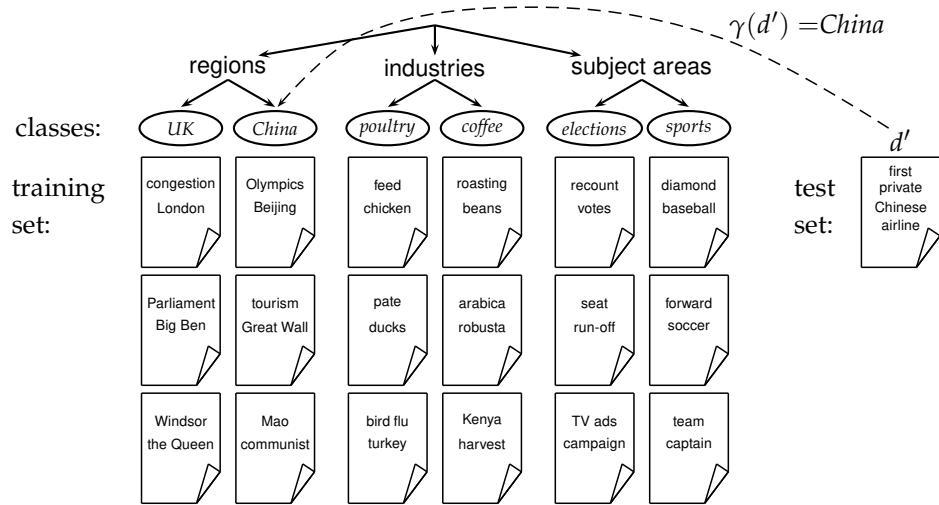
STATISTICAL TEXT
CLASSIFICATION

LABELING

sification as discussed in Section 13.5. Section 13.6 covers evaluation of text classification. In the following chapters, Chapters 14 and 15, we look at two other families of classification methods, vector space classifiers and support vector machines.

13.1 The text classification problem

DOCUMENT SPACE CLASS TRAINING SET LEARNING METHOD CLASSIFIER SUPervised LEARNING TEST SET	<p>In text classification, we are given a description $d \in \mathbb{X}$ of a document, where \mathbb{X} is the <i>document space</i>; and a fixed set of <i>classes</i> $\mathbb{C} = \{c_1, c_2, \dots, c_J\}$. Classes are also called <i>categories</i> or <i>labels</i>. Typically, the document space \mathbb{X} is some type of high-dimensional space, and the classes are human defined for the needs of an application, as in the examples <i>China</i> and <i>documents that talk about multicore computer chips</i> above. We are given a <i>training set</i> \mathbb{D} of labeled documents $\langle d, c \rangle$, where $\langle d, c \rangle \in \mathbb{X} \times \mathbb{C}$. For example:</p> $\langle d, c \rangle = \langle \text{Beijing joins the World Trade Organization}, \text{China} \rangle$ <p>for the one-sentence document <i>Beijing joins the World Trade Organization</i> and the class (or label) <i>China</i>.</p> <p>Using a <i>learning method</i> or <i>learning algorithm</i>, we then wish to learn a classifier or <i>classification function</i> γ that maps documents to classes:</p> $(13.1) \quad \gamma : \mathbb{X} \rightarrow \mathbb{C}$ <p>This type of learning is called <i>supervised learning</i> because a supervisor (the human who defines the classes and labels training documents) serves as a teacher directing the learning process. We denote the supervised learning method by Γ and write $\Gamma(\mathbb{D}) = \gamma$. The learning method Γ takes the training set \mathbb{D} as input and returns the learned classification function γ.</p> <p>Most names for learning methods Γ are also used for classifiers γ. We talk about the Naive Bayes (NB) <i>learning method</i> Γ when we say that “Naive Bayes is robust,” meaning that it can be applied to many different learning problems and is unlikely to produce classifiers that fail catastrophically. But when we say that “Naive Bayes had an error rate of 20%,” we are describing an experiment in which a particular NB <i>classifier</i> γ (which was produced by the NB learning method) had a 20% error rate in an application.</p> <p>Figure 13.1 shows an example of text classification from the Reuters-RCV1 collection, introduced in Section 4.2, page 69. There are six classes (<i>UK</i>, <i>China</i>, \dots, <i>sports</i>), each with three training documents. We show a few mnemonic words for each document’s content. The training set provides some typical examples for each class, so that we can learn the classification function γ. Once we have learned γ, we can apply it to the <i>test set</i> (or <i>test data</i>), for example, the new document <i>first private Chinese airline</i> whose class is unknown.</p>
-----------------------------------------------------------------------------------------------------------------------------	-----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------



► **Figure 13.1** Classes, training set, and test set in text classification .

In Figure 13.1, the classification function assigns the new document to class $\gamma(d) = \text{China}$, which is the correct assignment.

The classes in text classification often have some interesting structure such as the hierarchy in Figure 13.1. There are two instances each of region categories, industry categories, and subject area categories. A hierarchy can be an important aid in solving a classification problem; see Section 15.3.2 for further discussion. Until then, we will make the assumption in the text classification chapters that the classes form a set with no subset relationships between them.

Definition (13.1) stipulates that a document is a member of exactly one class. This is not the most appropriate model for the hierarchy in Figure 13.1. For instance, a document about the 2008 Olympics should be a member of two classes: the *China* class and the *sports* class. This type of classification problem is referred to as an *any-of* problem and we will return to it in Section 14.5 (page 306). For the time being, we only consider *one-of* problems where a document is a member of exactly one class.

Our goal in text classification is high accuracy on test data or *new data* – for example, the newswire articles that we will encounter tomorrow morning in the multicore chip example. It is easy to achieve high accuracy on the training set (e.g., we can simply memorize the labels). But high accuracy on the training set in general does not mean that the classifier will work well on

new data in an application. When we use the training set to learn a classifier for test data, we make the assumption that training data and test data are similar or from *the same distribution*. We defer a precise definition of this notion to Section 14.6 (page 308).

13.2 Naive Bayes text classification

MULTINOMIAL NAIVE
BAYES

The first supervised learning method we introduce is the *multinomial Naive Bayes* or *multinomial NB* model, a probabilistic learning method. The probability of a document d being in class c is computed as

$$(13.2) \quad P(c|d) \propto P(c) \prod_{1 \leq k \leq n_d} P(t_k|c)$$

where $P(t_k|c)$ is the conditional probability of term t_k occurring in a document of class c .¹ We interpret $P(t_k|c)$ as a measure of how much evidence t_k contributes that c is the correct class. $P(c)$ is the prior probability of a document occurring in class c . If a document's terms do not provide clear evidence for one class versus another, we choose the one that has a higher prior probability. $\langle t_1, t_2, \dots, t_{n_d} \rangle$ are the tokens in d that are part of the vocabulary we use for classification and n_d is the number of such tokens in d . For example, $\langle t_1, t_2, \dots, t_{n_d} \rangle$ for the one-sentence document *Beijing and Taipei join the WTO* might be $\langle \text{Beijing}, \text{Taipei}, \text{join}, \text{WTO} \rangle$, with $n_d = 4$, if we treat the terms and and the as stop words.

MAXIMUM A
POSTERIORI CLASS

In text classification, our goal is to find the *best* class for the document. The best class in NB classification is the most likely or *maximum a posteriori* (MAP) class c_{map} :

$$(13.3) \quad c_{\text{map}} = \arg \max_{c \in \mathcal{C}} \hat{P}(c|d) = \arg \max_{c \in \mathcal{C}} \hat{P}(c) \prod_{1 \leq k \leq n_d} \hat{P}(t_k|c).$$

We write \hat{P} for P because we do not know the true values of the parameters $P(c)$ and $P(t_k|c)$, but estimate them from the training set as we will see in a moment.

In Equation (13.3), many conditional probabilities are multiplied, one for each position $1 \leq k \leq n_d$. This can result in a floating point underflow. It is therefore better to perform the computation by adding logarithms of probabilities instead of multiplying probabilities. The class with the highest log probability score is still the most probable; $\log(xy) = \log(x) + \log(y)$ and the logarithm function is monotonic. Hence, the maximization that is

1. We will explain in the next section why $P(c|d)$ is proportional to (\propto) , not equal to the quantity on the right.

actually done in most implementations of NB is:

$$(13.4) \quad c_{\text{map}} = \arg \max_{c \in C} [\log \hat{P}(c) + \sum_{1 \leq k \leq n_d} \log \hat{P}(t_k|c)].$$

Equation (13.4) has a simple interpretation. Each conditional parameter $\log \hat{P}(t_k|c)$ is a weight that indicates how good an indicator t_k is for c . Similarly, the prior $\log \hat{P}(c)$ is a weight that indicates the relative frequency of c . More frequent classes are more likely to be the correct class than infrequent classes. The sum of log prior and term weights is then a measure of how much evidence there is for the document being in the class, and Equation (13.4) selects the class for which we have the most evidence.

We will initially work with this intuitive interpretation of the multinomial NB model and defer a formal derivation to Section 13.4.

How do we estimate the parameters $\hat{P}(c)$ and $\hat{P}(t_k|c)$? We first try the maximum likelihood estimate (MLE; Section 11.3.2, page 226), which is simply the relative frequency and corresponds to the most likely value of each parameter given the training data. For the priors this estimate is:

$$(13.5) \quad \hat{P}(c) = \frac{N_c}{N},$$

where N_c is the number of documents in class c and N is the total number of documents.

We estimate the conditional probability $\hat{P}(t|c)$ as the relative frequency of term t in documents belonging to class c :

$$(13.6) \quad \hat{P}(t|c) = \frac{T_{ct}}{\sum_{t' \in V} T_{ct'}},$$

where T_{ct} is the number of occurrences of t in training documents from class c , including multiple occurrences of a term in a document. We have made the *positional independence assumption* here, which we will discuss in more detail in the next section: T_{ct} is a count of occurrences in all positions k in the documents in the training set. Thus, we do not compute different estimates for different positions and, for example, if a word occurs twice in a document, in positions k_1 and k_2 , then $\hat{P}(t_{k_1}|c) = \hat{P}(t_{k_2}|c)$.

The problem with the MLE estimate is that it is zero for a term–class combination that did not occur in the training data. If the term WTO in the training data only occurred in *China* documents, then the MLE estimates for the other classes, for example UK, will be zero:

$$\hat{P}(\text{WTO}|UK) = 0.$$

Now, the one-sentence document *Britain is a member of the WTO* will get a conditional probability of zero for UK because we are multiplying the conditional probabilities for all terms in Equation (13.2). Clearly, the model should

```

TRAINMULTINOMIALNB( $\mathcal{C}, \mathbb{D}$ )
1  $V \leftarrow \text{EXTRACTVOCABULARY}(\mathbb{D})$ 
2  $N \leftarrow \text{COUNTDOCS}(\mathbb{D})$ 
3 for each  $c \in \mathcal{C}$ 
4   do  $N_c \leftarrow \text{COUNTDOCSINCLASS}(\mathbb{D}, c)$ 
5      $prior[c] \leftarrow N_c/N$ 
6      $text_c \leftarrow \text{CONCATENATETEXTOفالDOCSINCLASS}(\mathbb{D}, c)$ 
7     for each  $t \in V$ 
8       do  $T_{ct} \leftarrow \text{COUNTTOKENSOFTERM}(text_c, t)$ 
9       for each  $t \in V$ 
10      do  $condprob[t][c] \leftarrow \frac{T_{ct}+1}{\sum_{t'}(T_{ct'}+1)}$ 
11 return  $V, prior, condprob$ 

APPLYMULTINOMIALNB( $\mathcal{C}, V, prior, condprob, d$ )
1  $W \leftarrow \text{EXTRACTTOKENSFROMDOC}(V, d)$ 
2 for each  $c \in \mathcal{C}$ 
3   do  $score[c] \leftarrow \log prior[c]$ 
4     for each  $t \in W$ 
5       do  $score[c] += \log condprob[t][c]$ 
6 return  $\arg \max_{c \in \mathcal{C}} score[c]$ 

```

► **Figure 13.2** Naive Bayes algorithm (multinomial model): Training and testing.

SPARSENESS

assign a high probability to the *UK* class because the term *Britain* occurs. The problem is that the zero probability for *WTO* cannot be “conditioned away,” no matter how strong the evidence for the class *UK* from other features. The estimate is 0 because of *sparseness*: The training data are never large enough to represent the frequency of rare events adequately, for example, the frequency of *WTO* occurring in *UK* documents.

ADD-ONE SMOOTHING

To eliminate zeros, we use *add-one* or *Laplace smoothing*, which simply adds one to each count (cf. Section 11.3.2):

$$(13.7) \quad \hat{P}(t|c) = \frac{T_{ct} + 1}{\sum_{t' \in V} (T_{ct'} + 1)} = \frac{T_{ct} + 1}{(\sum_{t' \in V} T_{ct'}) + B},$$

where $B = |V|$ is the number of terms in the vocabulary. Add-one smoothing can be interpreted as a uniform prior (each term occurs once for each class) that is then updated as evidence from the training data comes in. Note that this is a prior probability for the occurrence of a *term* as opposed to the prior probability of a *class* which we estimate in Equation (13.5) on the document level.

► **Table 13.1** Data for parameter estimation examples.

	docID	words in document	in $c = China$?
training set	1	Chinese Beijing Chinese	yes
	2	Chinese Chinese Shanghai	yes
	3	Chinese Macao	yes
	4	Tokyo Japan Chinese	no
test set	5	Chinese Chinese Chinese Tokyo Japan	?

► **Table 13.2** Training and test times for NB.

mode	time complexity
training	$\Theta(\mathcal{D} L_{ave} + \mathcal{C} V)$
testing	$\Theta(L_a + \mathcal{C} M_a) = \Theta(\mathcal{C} M_a)$

We have now introduced all the elements we need for training and applying an NB classifier. The complete algorithm is described in Figure 13.2.

 **Example 13.1:** For the example in Table 13.1, the multinomial parameters we need to classify the test document are the priors $\hat{P}(c) = 3/4$ and $\hat{P}(\bar{c}) = 1/4$ and the following conditional probabilities:

$$\begin{aligned}\hat{P}(\text{Chinese}|c) &= (5+1)/(8+6) = 6/14 = 3/7 \\ \hat{P}(\text{Tokyo}|c) &= \hat{P}(\text{Japan}|c) = (0+1)/(8+6) = 1/14 \\ \hat{P}(\text{Chinese}|\bar{c}) &= (1+1)/(3+6) = 2/9 \\ \hat{P}(\text{Tokyo}|\bar{c}) &= \hat{P}(\text{Japan}|\bar{c}) = (1+1)/(3+6) = 2/9\end{aligned}$$

The denominators are $(8+6)$ and $(3+6)$ because the lengths of $text_c$ and $text_{\bar{c}}$ are 8 and 3, respectively, and because the constant B in Equation (13.7) is 6 as the vocabulary consists of six terms.

We then get:

$$\begin{aligned}\hat{P}(c|d_5) &\propto 3/4 \cdot (3/7)^3 \cdot 1/14 \cdot 1/14 \approx 0.0003. \\ \hat{P}(\bar{c}|d_5) &\propto 1/4 \cdot (2/9)^3 \cdot 2/9 \cdot 2/9 \approx 0.0001.\end{aligned}$$

Thus, the classifier assigns the test document to $c = China$. The reason for this classification decision is that the three occurrences of the positive indicator Chinese in d_5 outweigh the occurrences of the two negative indicators Japan and Tokyo.

What is the time complexity of NB? The complexity of computing the parameters is $\Theta(|\mathcal{C}||V|)$ because the set of parameters consists of $|\mathcal{C}||V|$ conditional probabilities and $|\mathcal{C}|$ priors. The preprocessing necessary for computing the parameters (extracting the vocabulary, counting terms, etc.) can be done in one pass through the training data. The time complexity of this

component is therefore $\Theta(|\mathbb{D}|L_{\text{ave}})$, where $|\mathbb{D}|$ is the number of documents and L_{ave} is the average length of a document.

We use $\Theta(|\mathbb{D}|L_{\text{ave}})$ as a notation for $\Theta(T)$ here, where T is the length of the training collection. This is nonstandard; $\Theta(\cdot)$ is not defined for an average. We prefer expressing the time complexity in terms of $|\mathbb{D}|$ and L_{ave} because these are the primary statistics used to characterize training collections.

The time complexity of `APPLYMULTINOMIALNB` in Figure 13.2 is $\Theta(|\mathbb{C}|L_a)$. L_a and M_a are the numbers of tokens and types, respectively, in the test document. `APPLYMULTINOMIALNB` can be modified to be $\Theta(L_a + |\mathbb{C}|M_a)$ (Exercise 13.8). Finally, assuming that the length of test documents is bounded, $\Theta(L_a + |\mathbb{C}|M_a) = \Theta(|\mathbb{C}|M_a)$ because $L_a < b|\mathbb{C}|M_a$ for a fixed constant b .²

Table 13.2 summarizes the time complexities. In general, we have $|\mathbb{C}||V| < |\mathbb{D}|L_{\text{ave}}$, so both training and testing complexity are linear in the time it takes to scan the data. Because we have to look at the data at least once, NB can be said to have optimal time complexity. Its efficiency is one reason why NB is a popular text classification method.

13.2.1 Relation to multinomial unigram language model

The multinomial NB model is formally identical to the multinomial unigram language model (Section 12.2.1, page 242). In particular, Equation (13.2) is a special case of Equation (12.12) from page 243, which we repeat here for $\lambda = 1$:

$$(13.8) \quad P(d|q) \propto P(d) \prod_{t \in q} P(t|M_d).$$

The document d in text classification (Equation (13.2)) takes the role of the query in language modeling (Equation (13.8)) and the classes c in text classification take the role of the documents d in language modeling. We used Equation (13.8) to rank documents according to the probability that they are relevant to the query q . In NB classification, we are usually only interested in the top-ranked class.

We also used MLE estimates in Section 12.2.2 (page 243) and encountered the problem of zero estimates owing to sparse data (page 244); but instead of add-one smoothing, we used a mixture of two distributions to address the problem there. Add-one smoothing is closely related to add- $\frac{1}{2}$ smoothing in Section 11.3.4 (page 228).



Exercise 13.1

Why is $|\mathbb{C}||V| < |\mathbb{D}|L_{\text{ave}}$ in Table 13.2 expected to hold for most text collections?

2. Our assumption here is that the length of test documents is bounded. L_a would exceed $b|\mathbb{C}|M_a$ for extremely long test documents.

```

TRAINBERNOULLINB( $\mathbb{C}, \mathbb{D}$ )
1  $V \leftarrow \text{EXTRACTVOCABULARY}(\mathbb{D})$ 
2  $N \leftarrow \text{COUNTDOCS}(\mathbb{D})$ 
3 for each  $c \in \mathbb{C}$ 
4 do  $N_c \leftarrow \text{COUNTDOCSINCLASS}(\mathbb{D}, c)$ 
5    $prior[c] \leftarrow N_c/N$ 
6   for each  $t \in V$ 
7   do  $N_{ct} \leftarrow \text{COUNTDOCSINCLASSCONTAININGTERM}(\mathbb{D}, c, t)$ 
8     $condprob[t][c] \leftarrow (N_{ct} + 1)/(N_c + 2)$ 
9 return  $V, prior, condprob$ 

APPLYBERNOULLINB( $\mathbb{C}, V, prior, condprob, d$ )
1  $V_d \leftarrow \text{EXTRACTTERMSFROMDOC}(V, d)$ 
2 for each  $c \in \mathbb{C}$ 
3 do  $score[c] \leftarrow \log prior[c]$ 
4   for each  $t \in V$ 
5   do if  $t \in V_d$ 
6     then  $score[c] += \log condprob[t][c]$ 
7     else  $score[c] += \log(1 - condprob[t][c])$ 
8 return  $\arg \max_{c \in \mathbb{C}} score[c]$ 

```

► **Figure 13.3** NB algorithm (Bernoulli model): Training and testing. The add-one smoothing in Line 8 (top) is in analogy to Equation (13.7) with $B = 2$.

13.3 The Bernoulli model

There are two different ways we can set up an NB classifier. The model we introduced in the previous section is the multinomial model. It generates one term from the vocabulary in each position of the document, where we assume a generative model that will be discussed in more detail in Section 13.4 (see also page 237).

BERNOULLI MODEL

An alternative to the multinomial model is the *multivariate Bernoulli model* or *Bernoulli model*. It is equivalent to the binary independence model of Section 11.3 (page 222), which generates an indicator for each term of the vocabulary, either 1 indicating presence of the term in the document or 0 indicating absence. Figure 13.3 presents training and testing algorithms for the Bernoulli model. The Bernoulli model has the same time complexity as the multinomial model.

The different generation models imply different estimation strategies and different classification rules. The Bernoulli model estimates $\hat{P}(t|c)$ as the fraction of documents of class c that contain term t (Figure 13.3, TRAINBERNOULLI-

NB, line 8). In contrast, the multinomial model estimates $\hat{P}(t|c)$ as the *fraction of tokens or fraction of positions* in documents of class c that contain term t (Equation (13.7)). When classifying a test document, the Bernoulli model uses binary occurrence information, ignoring the number of occurrences, whereas the multinomial model keeps track of multiple occurrences. As a result, the Bernoulli model typically makes many mistakes when classifying long documents. For example, it may assign an entire book to the class *China* because of a single occurrence of the term *China*.

The models also differ in how nonoccurring terms are used in classification. They do not affect the classification decision in the multinomial model; but in the Bernoulli model the probability of nonoccurrence is factored in when computing $P(c|d)$ (Figure 13.3, APPLYBERNOULLINB, Line 7). This is because only the Bernoulli NB model models absence of terms explicitly.

 **Example 13.2:** Applying the Bernoulli model to the example in Table 13.1, we have the same estimates for the priors as before: $\hat{P}(c) = 3/4$, $\hat{P}(\bar{c}) = 1/4$. The conditional probabilities are:

$$\begin{aligned}\hat{P}(\text{Chinese}|c) &= (3+1)/(3+2) = 4/5 \\ \hat{P}(\text{Japan}|c) = \hat{P}(\text{Tokyo}|c) &= (0+1)/(3+2) = 1/5 \\ \hat{P}(\text{Beijing}|c) = \hat{P}(\text{Macao}|c) = \hat{P}(\text{Shanghai}|c) &= (1+1)/(3+2) = 2/5 \\ \hat{P}(\text{Chinese}|\bar{c}) &= (1+1)/(1+2) = 2/3 \\ \hat{P}(\text{Japan}|\bar{c}) = \hat{P}(\text{Tokyo}|\bar{c}) &= (1+1)/(1+2) = 2/3 \\ \hat{P}(\text{Beijing}|\bar{c}) = \hat{P}(\text{Macao}|\bar{c}) = \hat{P}(\text{Shanghai}|\bar{c}) &= (0+1)/(1+2) = 1/3\end{aligned}$$

The denominators are $(3+2)$ and $(1+2)$ because there are three documents in c and one document in \bar{c} and because the constant B in Equation (13.7) is 2 – there are two cases to consider for each term, occurrence and nonoccurrence.

The scores of the test document for the two classes are

$$\begin{aligned}\hat{P}(c|d_5) &\propto \hat{P}(c) \cdot \hat{P}(\text{Chinese}|c) \cdot \hat{P}(\text{Japan}|c) \cdot \hat{P}(\text{Tokyo}|c) \\ &\quad \cdot (1 - \hat{P}(\text{Beijing}|c)) \cdot (1 - \hat{P}(\text{Shanghai}|c)) \cdot (1 - \hat{P}(\text{Macao}|c)) \\ &= 3/4 \cdot 4/5 \cdot 1/5 \cdot 1/5 \cdot (1-2/5) \cdot (1-2/5) \cdot (1-2/5) \\ &\approx 0.005\end{aligned}$$

and, analogously,

$$\begin{aligned}\hat{P}(\bar{c}|d_5) &\propto 1/4 \cdot 2/3 \cdot 2/3 \cdot 2/3 \cdot (1-1/3) \cdot (1-1/3) \cdot (1-1/3) \\ &\approx 0.022\end{aligned}$$

Thus, the classifier assigns the test document to $\bar{c} = \text{not-China}$. When looking only at binary occurrence and not at term frequency, Japan and Tokyo are indicators for \bar{c} ($2/3 > 1/5$) and the conditional probabilities of Chinese for c and \bar{c} are not different enough ($4/5$ vs. $2/3$) to affect the classification decision.

13.4 Properties of Naive Bayes

To gain a better understanding of the two models and the assumptions they make, let us go back and examine how we derived their classification rules in Chapters 11 and 12. We decide class membership of a document by assigning it to the class with the maximum a posteriori probability (cf. Section 11.3.2, page 226), which we compute as follows:

$$\begin{aligned} c_{\text{map}} &= \arg \max_{c \in \mathcal{C}} P(c|d) \\ (13.9) \quad &= \arg \max_{c \in \mathcal{C}} \frac{P(d|c)P(c)}{P(d)} \end{aligned}$$

$$(13.10) \quad = \arg \max_{c \in \mathcal{C}} P(d|c)P(c),$$

where Bayes' rule (Equation (11.4), page 220) is applied in (13.9) and we drop the denominator in the last step because $P(d)$ is the same for all classes and does not affect the argmax.

We can interpret Equation (13.10) as a description of the generative process we assume in Bayesian text classification. To generate a document, we first choose class c with probability $P(c)$ (top nodes in Figures 13.4 and 13.5). The two models differ in the formalization of the second step, the generation of the document given the class, corresponding to the conditional distribution $P(d|c)$:

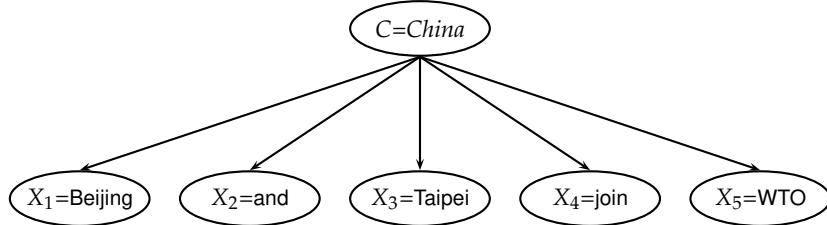
$$(13.11) \quad \text{Multinomial} \quad P(d|c) = P(\langle t_1, \dots, t_k, \dots, t_{n_d} \rangle | c)$$

$$(13.12) \quad \text{Bernoulli} \quad P(d|c) = P(\langle e_1, \dots, e_i, \dots, e_M \rangle | c),$$

where $\langle t_1, \dots, t_{n_d} \rangle$ is the sequence of terms as it occurs in d (minus terms that were excluded from the vocabulary) and $\langle e_1, \dots, e_i, \dots, e_M \rangle$ is a binary vector of dimensionality M that indicates for each term whether it occurs in d or not.

It should now be clearer why we introduced the document space \mathbb{X} in Equation (13.1) when we defined the classification problem. A critical step in solving a text classification problem is to choose the document representation. $\langle t_1, \dots, t_{n_d} \rangle$ and $\langle e_1, \dots, e_M \rangle$ are two different document representations. In the first case, \mathbb{X} is the set of all term sequences (or, more precisely, sequences of term tokens). In the second case, \mathbb{X} is $\{0, 1\}^M$.

We cannot use Equations (13.11) and (13.12) for text classification directly. For the Bernoulli model, we would have to estimate $2^M |\mathcal{C}|$ different parameters, one for each possible combination of M values e_i and a class. The number of parameters in the multinomial case has the same order of magni-



► Figure 13.4 The multinomial NB model.

tude.³ This being a very large quantity, estimating these parameters reliably is infeasible.

CONDITIONAL
INDEPENDENCE
ASSUMPTION

To reduce the number of parameters, we make the Naive Bayes *conditional independence assumption*. We assume that attribute values are independent of each other given the class:

$$(13.13) \quad \text{Multinomial} \quad P(d|c) = P(\langle t_1, \dots, t_{n_d} \rangle | c) = \prod_{1 \leq k \leq n_d} P(X_k = t_k | c)$$

$$(13.14) \quad \text{Bernoulli} \quad P(d|c) = P(\langle e_1, \dots, e_M \rangle | c) = \prod_{1 \leq i \leq M} P(U_i = e_i | c).$$

RANDOM VARIABLE X

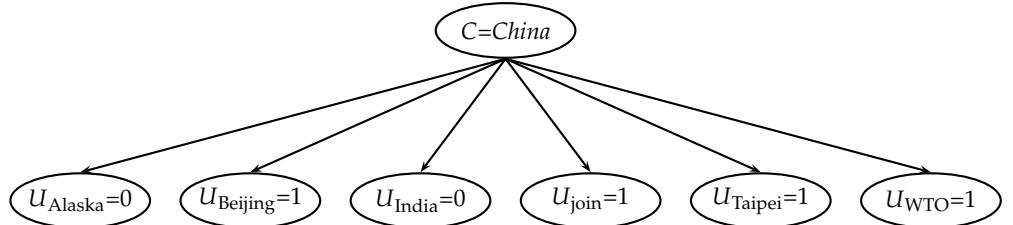
RANDOM VARIABLE U

We have introduced two random variables here to make the two different generative models explicit. X_k is the random variable for position k in the document and takes as values terms from the vocabulary. $P(X_k = t | c)$ is the probability that in a document of class c the term t will occur in position k . U_i is the random variable for vocabulary term i and takes as values 0 (absence) and 1 (presence). $P(U_i = 1 | c)$ is the probability that in a document of class c the term t_i will occur – in any position and possibly multiple times.

We illustrate the conditional independence assumption in Figures 13.4 and 13.5. The class *China* generates values for each of the five term attributes (multinomial) or six binary attributes (Bernoulli) with a certain probability, independent of the values of the other attributes. The fact that a document in the class *China* contains the term *Taipei* does not make it more likely or less likely that it also contains *Beijing*.

In reality, the conditional independence assumption does not hold for text data. Terms are conditionally dependent on each other. But as we will discuss shortly, NB models perform well despite the conditional independence assumption.

3. In fact, if the length of documents is not bounded, the number of parameters in the multinomial case is infinite.



► **Figure 13.5** The Bernoulli NB model.

Even when assuming conditional independence, we still have too many parameters for the multinomial model if we assume a different probability distribution for each position k in the document. The position of a term in a document by itself does not carry information about the class. Although there is a difference between *China sues France* and *France sues China*, the occurrence of *China* in position 1 versus position 3 of the document is not useful in NB classification because we look at each term separately. The conditional independence assumption commits us to this way of processing the evidence.

Also, if we assumed different term distributions for each position k , we would have to estimate a different set of parameters for each k . The probability of *bean* appearing as the first term of a *coffee* document could be different from it appearing as the second term, and so on. This again causes problems in estimation owing to data sparseness.

POSITIONAL
INDEPENDENCE

For these reasons, we make a second independence assumption for the multinomial model, *positional independence*: The conditional probabilities for a term are the same independent of position in the document.

$$P(X_{k_1} = t|c) = P(X_{k_2} = t|c)$$

for all positions k_1, k_2 , terms t and classes c . Thus, we have a single distribution of terms that is valid for all positions k_i and we can use X as its symbol.⁴ Positional independence is equivalent to adopting the bag of words model, which we introduced in the context of ad hoc retrieval in Chapter 6 (page 117).

With conditional and positional independence assumptions, we only need to estimate $\Theta(M|\mathcal{C}|)$ parameters $P(t_k|c)$ (multinomial model) or $P(e_i|c)$ (Bernoulli

4. Our terminology is nonstandard. The random variable X is a categorical variable, not a multinomial variable, and the corresponding NB model should perhaps be called a *sequence model*. We have chosen to present this sequence model and the multinomial model in Section 13.4.1 as the same model because they are computationally identical.

► **Table 13.3** Multinomial versus Bernoulli model.

	multinomial model	Bernoulli model
event model	generation of token	generation of document
random variable(s)	$X = t$ iff t occurs at given pos	$U_t = 1$ iff t occurs in doc
document representation	$d = \langle t_1, \dots, t_k, \dots, t_{n_d} \rangle, t_k \in V$	$d = \langle e_1, \dots, e_i, \dots, e_M \rangle, e_i \in \{0, 1\}$
parameter estimation	$\hat{P}(X = t c)$	$\hat{P}(U_i = e c)$
decision rule: maximize	$\hat{P}(c) \prod_{1 \leq k \leq n_d} \hat{P}(X = t_k c)$	$\hat{P}(c) \prod_{t_i \in V} \hat{P}(U_i = e_i c)$
multiple occurrences	taken into account	ignored
length of docs	can handle longer docs	works best for short docs
# features	can handle more	works best with fewer
estimate for term the	$\hat{P}(X = \text{the} c) \approx 0.05$	$\hat{P}(U_{\text{the}} = 1 c) \approx 1.0$

model), one for each term–class combination, rather than a number that is at least exponential in M , the size of the vocabulary. The independence assumptions reduce the number of parameters to be estimated by several orders of magnitude.

RANDOM VARIABLE C

To summarize, we generate a document in the multinomial model (Figure 13.4) by first picking a class $C = c$ with $P(c)$ where C is a random variable taking values from \mathcal{C} as values. Next we generate term t_k in position k with $P(X_k = t_k|c)$ for each of the n_d positions of the document. The X_k all have the same distribution over terms for a given c . In the example in Figure 13.4, we show the generation of $\langle t_1, t_2, t_3, t_4, t_5 \rangle = \langle \text{Beijing}, \text{and}, \text{Taipei}, \text{join}, \text{WTO} \rangle$, corresponding to the one-sentence document *Beijing and Taipei join WTO*.

For a completely specified document generation model, we would also have to define a distribution $P(n_d|c)$ over lengths. Without it, the multinomial model is a token generation model rather than a document generation model.

We generate a document in the Bernoulli model (Figure 13.5) by first picking a class $C = c$ with $P(c)$ and then generating a binary indicator e_i for each term t_i of the vocabulary ($1 \leq i \leq M$). In the example in Figure 13.5, we show the generation of $\langle e_1, e_2, e_3, e_4, e_5, e_6 \rangle = \langle 0, 1, 0, 1, 1, 1 \rangle$, corresponding, again, to the one-sentence document *Beijing and Taipei join WTO* where we have assumed that *and* is a stop word.

We compare the two models in Table 13.3, including estimation equations and decision rules.

Naive Bayes is so called because the independence assumptions we have just made are indeed very naive for a model of natural language. The conditional independence assumption states that features are independent of each other given the class. This is hardly ever true for terms in documents. In many cases, the opposite is true. The pairs *hong* and *kong* or *london* and *en-*

► **Table 13.4** Correct estimation implies accurate prediction, but accurate prediction does not imply correct estimation.

	c_1	c_2	class selected
true probability $P(c d)$	0.6	0.4	c_1
$\hat{P}(c) \prod_{1 \leq k \leq n_d} \hat{P}(t_k c)$ (Equation (13.13))	0.00099	0.00001	
NB estimate $\hat{P}(c d)$	0.99	0.01	c_1

glish in Figure 13.7 are examples of highly dependent terms. In addition, the multinomial model makes an assumption of positional independence. The Bernoulli model ignores positions in documents altogether because it only cares about absence or presence. This bag-of-words model discards all information that is communicated by the order of words in natural language sentences. How can NB be a good text classifier when its model of natural language is so oversimplified?

The answer is that even though the *probability estimates* of NB are of low quality, its *classification decisions* are surprisingly good. Consider a document d with true probabilities $P(c_1|d) = 0.6$ and $P(c_2|d) = 0.4$ as shown in Table 13.4. Assume that d contains many terms that are positive indicators for c_1 and many terms that are negative indicators for c_2 . Thus, when using the multinomial model in Equation (13.13), $\hat{P}(c_1) \prod_{1 \leq k \leq n_d} \hat{P}(t_k|c_1)$ will be much larger than $\hat{P}(c_2) \prod_{1 \leq k \leq n_d} \hat{P}(t_k|c_2)$ (0.00099 vs. 0.00001 in the table). After division by 0.001 to get well-formed probabilities for $P(c|d)$, we end up with one estimate that is close to 1.0 and one that is close to 0.0. This is common: The winning class in NB classification usually has a much larger probability than the other classes and the estimates diverge very significantly from the true probabilities. But the classification decision is based on which class gets the highest score. It does not matter how accurate the estimates are. Despite the bad estimates, NB estimates a higher probability for c_1 and therefore assigns d to the correct class in Table 13.4. *Correct estimation implies accurate prediction, but accurate prediction does not imply correct estimation.* NB classifiers estimate badly, but often classify well.

Even if it is not the method with the highest accuracy for text, NB has many virtues that make it a strong contender for text classification. It excels if there are many equally important features that jointly contribute to the classification decision. It is also somewhat robust to noise features (as defined in the next section) and *concept drift* – the gradual change over time of the concept underlying a class like *US president* from Bill Clinton to George W. Bush (see Section 13.7). Classifiers like kNN (Section 14.3, page 297) can be carefully tuned to idiosyncratic properties of a particular time period. This will then hurt them when documents in the following time period have slightly

► **Table 13.5** A set of documents for which the NB independence assumptions are problematic.

- (1) He moved from London, Ontario, to London, England.
 - (2) He moved from London, England, to London, Ontario.
 - (3) He moved from England to London, Ontario.
-

different properties.

The Bernoulli model is particularly robust with respect to concept drift. We will see in Figure 13.8 that it can have decent performance when using fewer than a dozen terms. The most important indicators for a class are less likely to change. Thus, a model that only relies on these features is more likely to maintain a certain level of accuracy in concept drift.

NB's main strength is its efficiency: Training and classification can be accomplished with one pass over the data. Because it combines efficiency with good accuracy it is often used as a baseline in text classification research. It is often the method of choice if (i) squeezing out a few extra percentage points of accuracy is not worth the trouble in a text classification application, (ii) a very large amount of training data is available and there is more to be gained from training on a lot of data than using a better classifier on a smaller training set, or (iii) if its robustness to concept drift can be exploited.

In this book, we discuss NB as a classifier for text. The independence assumptions do not hold for text. However, it can be shown that NB is an *optimal classifier* (in the sense of minimal error rate on new data) for data where the independence assumptions do hold.

OPTIMAL CLASSIFIER

13.4.1 A variant of the multinomial model

An alternative formalization of the multinomial model represents each document d as an M -dimensional vector of counts $\langle \text{tf}_{t_1,d}, \dots, \text{tf}_{t_M,d} \rangle$ where $\text{tf}_{t_i,d}$ is the term frequency of t_i in d . $P(d|c)$ is then computed as follows (cf. Equation (12.8), page 243);

$$(13.15) \quad P(d|c) = P(\langle \text{tf}_{t_1,d}, \dots, \text{tf}_{t_M,d} \rangle | c) \propto \prod_{1 \leq i \leq M} P(X = t_i | c)^{\text{tf}_{t_i,d}}$$

Note that we have omitted the multinomial factor. See Equation (12.8) (page 243).

Equation (13.15) is equivalent to the sequence model in Equation (13.2) as $P(X = t_i | c)^{\text{tf}_{t_i,d}} = 1$ for terms that do not occur in d ($\text{tf}_{t_i,d} = 0$) and a term that occurs $\text{tf}_{t_i,d} \geq 1$ times will contribute $\text{tf}_{t_i,d}$ factors both in Equation (13.2) and in Equation (13.15).

```

SELECTFEATURES( $\mathbb{D}, c, k$ )
1  $V \leftarrow \text{EXTRACTVOCABULARY}(\mathbb{D})$ 
2  $L \leftarrow []$ 
3 for each  $t \in V$ 
4 do  $A(t, c) \leftarrow \text{COMPUTEFEATUREUTILITY}(\mathbb{D}, t, c)$ 
5 APPEND( $L, \langle A(t, c), t \rangle$ )
6 return FEATURESWITHLARGESTVALUES( $L, k$ )

```

► **Figure 13.6** Basic feature selection algorithm for selecting the k best features.



Exercise 13.2

[*]

Which of the documents in Table 13.5 have identical and different bag of words representations for (i) the Bernoulli model (ii) the multinomial model? If there are differences, describe them.

Exercise 13.3

The rationale for the positional independence assumption is that there is no useful information in the fact that a term occurs in position k of a document. Find exceptions. Consider formulaic documents with a fixed document structure.

Exercise 13.4

Table 13.3 gives Bernoulli and multinomial estimates for the word *the*. Explain the difference.

13.5 Feature selection

FEATURE SELECTION

Feature selection is the process of selecting a subset of the terms occurring in the training set and using only this subset as features in text classification. Feature selection serves two main purposes. First, it makes training and applying a classifier more efficient by decreasing the size of the effective vocabulary. This is of particular importance for classifiers that, unlike NB, are expensive to train. Second, feature selection often increases classification accuracy by eliminating noise features. A *noise feature* is one that, when added to the document representation, increases the classification error on new data. Suppose a rare term, say *arachnocentric*, has no information about a class, say *China*, but all instances of *arachnocentric* happen to occur in *China* documents in our training set. Then the learning method might produce a classifier that misassigns test documents containing *arachnocentric* to *China*. Such an incorrect generalization from an accidental property of the training set is called *overfitting*.

NOISE FEATURE

We can view feature selection as a method for replacing a complex classifier (using all features) with a simpler one (using a subset of the features).

OVERFITTING

It may appear counterintuitive at first that a seemingly weaker classifier is advantageous in statistical text classification, but when discussing the bias-variance tradeoff in Section 14.6 (page 308), we will see that weaker models are often preferable when limited training data are available.

The basic feature selection algorithm is shown in Figure 13.6. For a given class c , we compute a utility measure $A(t, c)$ for each term of the vocabulary and select the k terms that have the highest values of $A(t, c)$. All other terms are discarded and not used in classification. We will introduce three different utility measures in this section: mutual information, $A(t, c) = I(U_t; C_c)$; the χ^2 test, $A(t, c) = \chi^2(t, c)$; and frequency, $A(t, c) = N(t, c)$.

Of the two NB models, the Bernoulli model is particularly sensitive to noise features. A Bernoulli NB classifier requires some form of feature selection or else its accuracy will be low.

This section mainly addresses feature selection for two-class classification tasks like *China* versus *not-China*. Section 13.5.5 briefly discusses optimizations for systems with more than two classes.

13.5.1 Mutual information

MUTUAL INFORMATION

A common feature selection method is to compute $A(t, c)$ as the expected *mutual information* (MI) of term t and class c .⁵ MI measures how much information the presence/absence of a term contributes to making the correct classification decision on c . Formally:

$$(13.16) \quad I(U; C) = \sum_{e_t \in \{1, 0\}} \sum_{e_c \in \{1, 0\}} P(U = e_t, C = e_c) \log_2 \frac{P(U = e_t, C = e_c)}{P(U = e_t)P(C = e_c)},$$

where U is a random variable that takes values $e_t = 1$ (the document contains term t) and $e_t = 0$ (the document does not contain t), as defined on page 266, and C is a random variable that takes values $e_c = 1$ (the document is in class c) and $e_c = 0$ (the document is not in class c). We write U_t and C_c if it is not clear from context which term t and class c we are referring to.

For MLEs of the probabilities, Equation (13.16) is equivalent to Equation (13.17):

$$(13.17) \quad I(U; C) = \frac{N_{11}}{N} \log_2 \frac{NN_{11}}{N_{1.}N_{.1}} + \frac{N_{01}}{N} \log_2 \frac{NN_{01}}{N_{0.}N_{.1}} \\ + \frac{N_{10}}{N} \log_2 \frac{NN_{10}}{N_{1.}N_{0.}} + \frac{N_{00}}{N} \log_2 \frac{NN_{00}}{N_{0.}N_{0.}}$$

where the N s are counts of documents that have the values of e_t and e_c that are indicated by the two subscripts. For example, N_{10} is the number of doc-

5. Take care not to confuse expected mutual information with *pointwise mutual information*, which is defined as $\log N_{11}/E_{11}$ where N_{11} and E_{11} are defined as in Equation (13.18). The two measures have different properties. See Section 13.7.

uments that contain t ($e_t = 1$) and are not in c ($e_c = 0$). $N_{1.} = N_{10} + N_{11}$ is the number of documents that contain t ($e_t = 1$) and we count documents independent of class membership ($e_c \in \{0, 1\}$). $N = N_{00} + N_{01} + N_{10} + N_{11}$ is the total number of documents. An example of one of the MLE estimates that transform Equation (13.16) into Equation (13.17) is $P(U = 1, C = 1) = N_{11}/N$.



Example 13.3: Consider the class *poultry* and the term *export* in Reuters-RCV1. The counts of the number of documents with the four possible combinations of indicator values are as follows:

	$e_c = e_{\text{poultry}} = 1$	$e_c = e_{\text{poultry}} = 0$
$e_t = e_{\text{export}} = 1$	$N_{11} = 49$	$N_{10} = 27,652$
$e_t = e_{\text{export}} = 0$	$N_{01} = 141$	$N_{00} = 774,106$

After plugging these values into Equation (13.17) we get:

$$\begin{aligned}
 I(U; C) &= \frac{49}{801,948} \log_2 \frac{801,948 \cdot 49}{(49+27,652)(49+141)} \\
 &\quad + \frac{141}{801,948} \log_2 \frac{801,948 \cdot 141}{(141+774,106)(49+141)} \\
 &\quad + \frac{27,652}{801,948} \log_2 \frac{801,948 \cdot 27,652}{(49+27,652)(27,652+774,106)} \\
 &\quad + \frac{774,106}{801,948} \log_2 \frac{801,948 \cdot 774,106}{(141+774,106)(27,652+774,106)} \\
 &\approx 0.0001105
 \end{aligned}$$

To select k terms t_1, \dots, t_k for a given class, we use the feature selection algorithm in Figure 13.6: We compute the utility measure as $A(t, c) = I(U_t, C_c)$ and select the k terms with the largest values.

Mutual information measures how much information – in the information-theoretic sense – a term contains about the class. If a term's distribution is the same in the class as it is in the collection as a whole, then $I(U; C) = 0$. MI reaches its maximum value if the term is a perfect indicator for class membership, that is, if the term is present in a document if and only if the document is in the class.

Figure 13.7 shows terms with high mutual information scores for the six classes in Figure 13.1.⁶ The selected terms (e.g., *london*, *uk*, *british* for the class *UK*) are of obvious utility for making classification decisions for their respective classes. At the bottom of the list for *UK* we find terms like *peripherals* and *tonight* (not shown in the figure) that are clearly not helpful in deciding

6. Feature scores were computed on the first 100,000 documents, except for *poultry*, a rare class, for which 800,000 documents were used. We have omitted numbers and other special words from the top ten lists.

<i>UK</i>		<i>China</i>		<i>poultry</i>	
london	0.1925	china	0.0997	poultry	0.0013
uk	0.0755	chinese	0.0523	meat	0.0008
british	0.0596	beijing	0.0444	chicken	0.0006
stg	0.0555	yuan	0.0344	agriculture	0.0005
britain	0.0469	shanghai	0.0292	avian	0.0004
plc	0.0357	hong	0.0198	broiler	0.0003
england	0.0238	kong	0.0195	veterinary	0.0003
pence	0.0212	xinhua	0.0155	birds	0.0003
pounds	0.0149	province	0.0117	inspection	0.0003
english	0.0126	taiwan	0.0108	pathogenic	0.0003

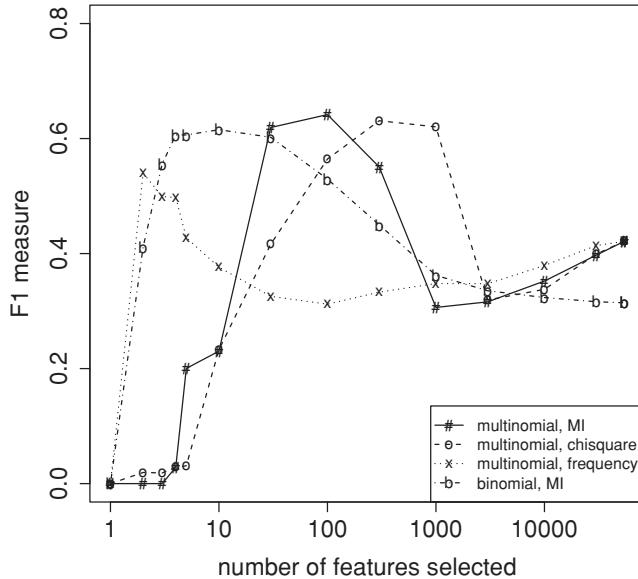
<i>coffee</i>		<i>elections</i>		<i>sports</i>	
coffee	0.0111	election	0.0519	soccer	0.0681
bags	0.0042	elections	0.0342	cup	0.0515
growers	0.0025	polls	0.0339	match	0.0441
kg	0.0019	voters	0.0315	matches	0.0408
colombia	0.0018	party	0.0303	played	0.0388
brazil	0.0016	vote	0.0299	league	0.0386
export	0.0014	poll	0.0225	beat	0.0301
exporters	0.0013	candidate	0.0202	game	0.0299
exports	0.0013	campaign	0.0202	games	0.0284
crop	0.0012	democratic	0.0198	team	0.0264

► **Figure 13.7** Features with high mutual information scores for six Reuters-RCV1 classes.

whether the document is in the class. As you might expect, keeping the informative terms and eliminating the non-informative ones tends to reduce noise and improve the classifier’s accuracy.

Such an accuracy increase can be observed in Figure 13.8, which shows F_1 as a function of vocabulary size after feature selection for Reuters-RCV1.⁷ Comparing F_1 at 132,776 features (corresponding to selection of all features) and at 10–100 features, we see that MI feature selection increases F_1 by about 0.1 for the multinomial model and by more than 0.2 for the Bernoulli model. For the Bernoulli model, F_1 peaks early, at ten features selected. At that point, the Bernoulli model is better than the multinomial model. When basing a classification decision on only a few features, it is more robust to consider binary occurrence only. For the multinomial model (MI feature selection), the peak occurs later, at 100 features, and its effectiveness recovers somewhat at

7. We trained the classifiers on the first 100,000 documents and computed F_1 on the next 100,000. The graphs are averages over five classes.



► **Figure 13.8** Effect of feature set size on accuracy for multinomial and Bernoulli models.

the end when we use all features. The reason is that the multinomial takes the number of occurrences into account in parameter estimation and classification and therefore better exploits a larger number of features than the Bernoulli model. Regardless of the differences between the two methods, using a carefully selected subset of the features results in better effectiveness than using all features.

13.5.2 χ^2 Feature selection

χ^2 FEATURE SELECTION

INDEPENDENCE

Another popular feature selection method is χ^2 . In statistics, the χ^2 test is applied to test the independence of two events, where two events A and B are defined to be *independent* if $P(AB) = P(A)P(B)$ or, equivalently, $P(A|B) = P(A)$ and $P(B|A) = P(B)$. In feature selection, the two events are occurrence of the term and occurrence of the class. We then rank terms with respect to the following quantity:

$$(13.18) \quad X^2(\mathbb{D}, t, c) = \sum_{e_t \in \{0,1\}} \sum_{e_c \in \{0,1\}} \frac{(N_{e_t e_c} - E_{e_t e_c})^2}{E_{e_t e_c}}$$

where e_t and e_c are defined as in Equation (13.16). N is the *observed* frequency in \mathbb{D} and E the *expected* frequency. For example, E_{11} is the expected frequency of t and c occurring together in a document assuming that term and class are independent.

 **Example 13.4:** We first compute E_{11} for the data in Example 13.3:

$$\begin{aligned} E_{11} &= N \times P(t) \times P(c) = N \times \frac{N_{11} + N_{10}}{N} \times \frac{N_{11} + N_{01}}{N} \\ &= N \times \frac{49 + 141}{N} \times \frac{49 + 27652}{N} \approx 6.6 \end{aligned}$$

where N is the total number of documents as before.

We compute the other $E_{e_t e_c}$ in the same way:

	$e_{poultry} = 1$	$e_{poultry} = 0$
$e_{\text{export}} = 1$	$N_{11} = 49$ $E_{11} \approx 6.6$	$N_{10} = 27,652$ $E_{10} \approx 27,694.4$
$e_{\text{export}} = 0$	$N_{01} = 141$ $E_{01} \approx 183.4$	$N_{00} = 774,106$ $E_{00} \approx 774,063.6$

Plugging these values into Equation (13.18), we get a X^2 value of 284:

$$X^2(\mathbb{D}, t, c) = \sum_{e_t \in \{0,1\}} \sum_{e_c \in \{0,1\}} \frac{(N_{e_t e_c} - E_{e_t e_c})^2}{E_{e_t e_c}} \approx 284$$

STATISTICAL
SIGNIFICANCE

X^2 is a measure of how much expected counts E and observed counts N deviate from each other. A high value of X^2 indicates that the hypothesis of independence, which implies that expected and observed counts are similar, is incorrect. In our example, $X^2 \approx 284 > 10.83$. Based on Table 13.6, we can reject the hypothesis that *poultry* and *export* are independent with only a 0.001 chance of being wrong.⁸ Equivalently, we say that the outcome $X^2 \approx 284 > 10.83$ is *statistically significant* at the 0.001 level. If the two events are dependent, then the occurrence of the term makes the occurrence of the class more likely (or less likely), so it should be helpful as a feature. This is the rationale of χ^2 feature selection.

An arithmetically simpler way of computing X^2 is the following:

$$(13.19) \quad X^2(\mathbb{D}, t, c) = \frac{(N_{11} + N_{10} + N_{01} + N_{00}) \times (N_{11}N_{00} - N_{10}N_{01})^2}{(N_{11} + N_{01}) \times (N_{11} + N_{10}) \times (N_{10} + N_{00}) \times (N_{01} + N_{00})}$$

This is equivalent to Equation (13.18) (Exercise 13.14).

⁸. We can make this inference because, if the two events are independent, then $X^2 \sim \chi^2$, where χ^2 is the χ^2 distribution. See, for example, Rice (2006).

► **Table 13.6** Critical values of the χ^2 distribution with one degree of freedom. For example, if the two events are independent, then $P(X^2 > 6.63) < 0.01$. So for $X^2 > 6.63$ the assumption of independence can be rejected with 99% confidence.

p	χ^2 critical value
0.1	2.71
0.05	3.84
0.01	6.63
0.005	7.88
0.001	10.83



Assessing χ^2 as a feature selection method

From a statistical point of view, χ^2 feature selection is problematic. For a test with one degree of freedom, the so-called Yates correction should be used (see Section 13.7), which makes it harder to reach statistical significance. Also, whenever a statistical test is used multiple times, then the probability of getting at least one error increases. If 1,000 hypotheses are rejected, each with 0.05 error probability, then $0.05 \times 1000 = 50$ calls of the test will be wrong on average. However, in text classification it rarely matters whether a few additional terms are added to the feature set or removed from it. Rather, the *relative* importance of features is important. As long as χ^2 feature selection only ranks features with respect to their usefulness and is not used to make statements about statistical dependence or independence of variables, we need not be overly concerned that it does not adhere strictly to statistical theory.

13.5.3 Frequency-based feature selection

A third feature selection method is *frequency-based feature selection*, that is, selecting the terms that are most common in the class. Frequency can be either defined as document frequency (the number of documents in the class c that contain the term t) or as collection frequency (the number of tokens of t that occur in documents in c). Document frequency is more appropriate for the Bernoulli model, collection frequency for the multinomial model.

Frequency-based feature selection selects some frequent terms that have no specific information about the class, for example, the days of the week (Monday, Tuesday, ...), which are frequent across classes in newswire text. When many thousands of features are selected, then frequency-based feature selection often does well. Thus, if somewhat suboptimal accuracy is acceptable, then frequency-based feature selection can be a good alternative to more complex methods. However, Figure 13.8 is a case where frequency-

based feature selection performs a lot worse than MI and χ^2 and should not be used.

13.5.4 Feature selection for multiple classifiers

In an operational system with a large number of classifiers, it is desirable to select a single set of features instead of a different one for each classifier. One way of doing this is to compute the X^2 statistic for an $n \times 2$ table where the columns are occurrence and nonoccurrence of the term and each row corresponds to one of the classes. We can then select the k terms with the highest X^2 statistic as before.

More commonly, feature selection statistics are first computed separately for each class on the two-class classification task c versus \bar{c} and then combined. One combination method computes a single figure of merit for each feature, for example, by averaging the values $A(t, c)$ for feature t , and then selects the k features with highest figures of merit. Another frequently used combination method selects the top k/n features for each of n classifiers and then combines these n sets into one global feature set.

Classification accuracy often decreases when selecting k common features for a system with n classifiers as opposed to n different sets of size k . But even if it does, the gain in efficiency owing to a common document representation may be worth the loss in accuracy.

13.5.5 Comparison of feature selection methods

Mutual information and χ^2 represent rather different feature selection methods. The independence of term t and class c can sometimes be rejected with high confidence even if t carries little information about membership of a document in c . This is particularly true for rare terms. If a term occurs once in a large collection and that one occurrence is in the *poultry* class, then this is statistically significant. But a single occurrence is not very informative according to the information-theoretic definition of information. Because its criterion is significance, χ^2 selects more rare terms (which are often less reliable indicators) than mutual information. But the selection criterion of mutual information also does not necessarily select the terms that maximize classification accuracy.

Despite the differences between the two methods, the classification accuracy of feature sets selected with χ^2 and MI does not seem to differ systematically. In most text classification problems, there are a few strong indicators and many weak indicators. As long as all strong indicators and a large number of weak indicators are selected, accuracy is expected to be good. Both methods do this.

Figure 13.8 compares MI and χ^2 feature selection for the multinomial model.

Peak effectiveness is virtually the same for both methods. χ^2 reaches this peak later, at 300 features, probably because the rare, but highly significant features it selects initially do not cover all documents in the class. However, features selected later (in the range of 100–300) are of better quality than those selected by MI.

GREEDY FEATURE SELECTION

All three methods – MI, χ^2 and frequency based – are *greedy* methods. They may select features that contribute no incremental information over previously selected features. In Figure 13.7, kong is selected as the seventh term even though it is highly correlated with previously selected hong and therefore redundant. Although such redundancy can negatively impact accuracy, non-greedy methods (see Section 13.7 for references) are rarely used in text classification due to their computational cost.



Exercise 13.5

Consider the following frequencies for the class *coffee* for four terms in the first 100,000 documents of Reuters-RCV1:

term	N_{00}	N_{01}	N_{10}	N_{11}
brazil	98,012	102	1835	51
council	96,322	133	3525	20
producers	98,524	119	1118	34
roasted	99,824	143	23	10

Select two of these four terms based on (i) χ^2 , (ii) mutual information, (iii) frequency.

13.6 Evaluation of text classification

TWO-CLASS CLASSIFIER

] Historically, the classic Reuters-21578 collection was the main benchmark for text classification evaluation. This is a collection of 21,578 newswire articles, originally collected and labeled by Carnegie Group, Inc. and Reuters, Ltd. in the course of developing the CONSTRUE text classification system. It is much smaller than and predates the Reuters-RCV1 collection discussed in Chapter 4 (page 69). The articles are assigned classes from a set of 118 topic categories. A document may be assigned several classes or none, but the commonest case is single assignment (documents with at least one class received an average of 1.24 classes). The standard approach to this *any-of* problem (Chapter 14, page 306) is to learn 118 two-class classifiers, one for each class, where the *two-class classifier* for class c is the classifier for the two classes c and its complement \bar{c} .

MODAPTE SPLIT

For each of these classifiers, we can measure recall, precision, and accuracy. In recent work, people almost invariably use the *ModApte split*, which includes only documents that were viewed and assessed by a human indexer,

► **Table 13.7** The ten largest classes in the Reuters-21578 collection with number of documents in training and test sets.

class	# train	# test	class	# train	# test
<i>earn</i>	2877	1087	<i>trade</i>	369	119
<i>acquisitions</i>	1650	179	<i>interest</i>	347	131
<i>money-fx</i>	538	179	<i>ship</i>	197	89
<i>grain</i>	433	149	<i>wheat</i>	212	71
<i>crude</i>	389	189	<i>corn</i>	182	56

and comprises 9,603 training documents and 3,299 test documents. The distribution of documents in classes is very uneven, and some work evaluates systems on only documents in the ten largest classes. They are listed in Table 13.7. A typical document with topics is shown in Figure 13.9.

In Section 13.1, we stated as our goal in text classification the minimization of classification error on test data. Classification error is 1.0 minus classification accuracy, the proportion of correct decisions, a measure we introduced in Section 8.3 (page 155). This measure is appropriate if the percentage of documents in the class is high, perhaps 10% to 20% and higher. But as we discussed in Section 8.3, accuracy is not a good measure for “small” classes because always saying no, a strategy that defeats the purpose of building a classifier, will achieve high accuracy. The always-no classifier is 99% accurate for a class with relative frequency 1%. For small classes, precision, recall and F_1 are better measures.

EFFECTIVENESS

We will use *effectiveness* as a generic term for measures that evaluate the quality of classification decisions, including precision, recall, F_1 , and accuracy. *Performance* refers to the computational efficiency of classification and IR systems in this book. However, many researchers mean effectiveness, not efficiency of text classification when they use the term performance.

MACROAVERAGING MICROAVERAGING

When we process a collection with several two-class classifiers (such as Reuters-21578 with its 118 classes), we often want to compute a single aggregate measure that combines the measures for individual classifiers. There are two methods for doing this. *Macroaveraging* computes a simple average over classes. *Microaveraging* pools per-document decisions across classes, and then computes an effectiveness measure on the pooled contingency table. Table 13.8 gives an example.

The differences between the two methods can be large. Macroaveraging gives equal weight to each class, whereas microaveraging gives equal weight to each per-document classification decision. Because the F_1 measure ignores true negatives and its magnitude is mostly determined by the number of true positives, large classes dominate small classes in microaveraging. In the example, microaveraged precision (0.83) is much closer to the precision of

```

<REUTERS TOPICS='YES' LEWISSPLIT='TRAIN'
CGISPLIT='TRAINING-SET' OLDDID='12981' NEWID='798'>
<DATE> 2-MAR-1987 16:51:43.42</DATE>
<TOPICS><D>livestock</D><D>hog</D></TOPICS>
<TITLE>AMERICAN PORK CONGRESS KICKS OFF TOMORROW</TITLE>
<DATELINE> CHICAGO, March 2 - </DATELINE><BODY>The American Pork
Congress kicks off tomorrow, March 3, in Indianapolis with 160
of the nations pork producers from 44 member states determining
industry positions on a number of issues, according to the
National Pork Producers Council, NPPC.
Delegates to the three day Congress will be considering 26
resolutions concerning various issues, including the future
direction of farm policy and the tax law as it applies to the
agriculture sector. The delegates will also debate whether to
endorse concepts of a national PRV (pseudorabies virus) control
and eradication program, the NPPC said. A large
trade show, in conjunction with the congress, will feature
the latest in technology in all areas of the industry, the NPPC
added. Reuter
\&\#3; </BODY></TEXT></REUTERS>
```

► **Figure 13.9** A sample document from the Reuters-21578 collection.

c_2 (0.9) than to the precision of c_1 (0.5) because c_2 is five times larger than c_1 . Microaveraged results are therefore really a measure of effectiveness on the large classes in a test collection. To get a sense of effectiveness on small classes, you should compute macroaveraged results.

In one-of classification (Section 14.5, page 306), microaveraged F_1 is the same as accuracy (Exercise 13.6).

Table 13.9 gives microaveraged and macroaveraged effectiveness of Naive Bayes for the ModApte split of Reuters-21578. To give a sense of the relative effectiveness of NB, we compare it with linear SVMs (rightmost column; see Chapter 15), one of the most effective classifiers, but also one that is more expensive to train than NB. NB has a microaveraged F_1 of 80%, which is 9% less than the SVM (89%), a 10% relative decrease (row “micro-avg-L (90 classes)”). So there is a surprisingly small effectiveness penalty for its simplicity and efficiency. However, on small classes, some of which only have on the order of ten positive examples in the training set, NB does much worse. Its macroaveraged F_1 is 13% below the SVM, a 22% relative decrease (row “macro-avg (90 classes)”).

The table also compares NB with the other classifiers we cover in this book:

► **Table 13.8** Macro- and microaveraging. “Truth” is the true class and “call” the decision of the classifier. In this example, macroaveraged precision is $[10/(10+10) + 90/(10+90)]/2 = (0.5+0.9)/2 = 0.7$. Microaveraged precision is $100/(100+20) \approx 0.83$.

class 1		class 2		pooled table	
truth:	truth:	truth:	truth:	truth:	truth:
yes	no	yes	no	yes	no
call: yes	10	10	call: yes	90	10
call: no	10	970	call: no	10	890
				call: yes	100
				call: no	20
					1860

► **Table 13.9** Text classification effectiveness numbers on Reuters-21578 for F_1 (in percent). Results from Li and Yang (2003) (a), Joachims (1998) (b: kNN) and Dumais et al. (1998) (b: NB, Rocchio, trees, SVM).

(a)	NB	Rocchio	kNN	SVM	
micro-avg-L (90 classes)	80	85	86	89	
macro-avg (90 classes)	47	59	60	60	
(b)	NB	Rocchio	kNN	trees	SVM
earn	96	93	97	98	98
acq	88	65	92	90	94
money-fx	57	47	78	66	75
grain	79	68	82	85	95
crude	80	70	86	85	89
trade	64	65	77	73	76
interest	65	63	74	67	78
ship	85	49	79	74	86
wheat	70	69	77	93	92
corn	65	48	78	92	90
micro-avg (top 10)	82	65	82	88	92
micro-avg-D (118 classes)	75	62	n/a	n/a	87

DECISION TREES

Rocchio and kNN. In addition, we give numbers for *decision trees*, an important classification method we do not cover. The bottom part of the table shows that there is considerable variation from class to class. For instance, NB beats kNN on *ship*, but is much worse on *money-fx*.

Comparing parts (a) and (b) of the table, one is struck by the degree to which the cited papers’ results differ. This is partly due to the fact that the numbers in (b) are break-even scores (cf. page 161) averaged over 118 classes, whereas the numbers in (a) are true F_1 scores (computed without any know-

ledge of the test set) averaged over ninety classes. This is unfortunately typical of what happens when comparing different results in text classification: There are often differences in the experimental setup or the evaluation that complicate the interpretation of the results.

These and other results have shown that the average effectiveness of NB is uncompetitive with classifiers like SVMs when trained and tested on *independent and identically distributed (i.i.d.)* data, that is, uniform data with all the good properties of statistical sampling. However, these differences may often be invisible or even reverse themselves when working in the real world where, usually, the training sample is drawn from a subset of the data to which the classifier will be applied, the nature of the data drifts over time rather than being stationary (the problem of concept drift we mentioned on page 269), and there may well be errors in the data (among other problems). Many practitioners have had the experience of being unable to build a fancy classifier for a certain problem that consistently performs better than NB.

Our conclusion from the results in Table 13.9 is that, although most researchers believe that an SVM is better than kNN and kNN better than NB, the ranking of classifiers ultimately depends on the class, the document collection, and the experimental setup. In text classification, there is always more to know than simply which machine learning algorithm was used, as we further discuss in Section 15.3 (page 334).

When performing evaluations like the one in Table 13.9, it is important to maintain a strict separation between the training set and the test set. We can easily make correct classification decisions on the test set by using information we have gleaned from the test set, such as the fact that a particular term is a good predictor in the test set (even though this is not the case in the training set). A more subtle example of using knowledge about the test set is to try a large number of values of a parameter (e.g., the number of selected features) and select the value that is best for the test set. As a rule, accuracy on new data – the type of data we will encounter when we use the classifier in an application – will be much lower than accuracy on a test set that the classifier has been tuned for. We discussed the same problem in ad hoc retrieval in Section 8.1 (page 153).

In a clean statistical text classification experiment, you should never run any program on or even look at the test set while developing a text classification system. Instead, set aside a *development set* for testing while you develop your method. When such a set serves the primary purpose of finding a good value for a parameter, for example, the number of selected features, then it is also called *held-out data*. Train the classifier on the rest of the training set with different parameter values, and then select the value that gives best results on the held-out part of the training set. Ideally, at the very end, when all parameters have been set and the method is fully specified, you run one final experiment on the test set and publish the results. Because no informa-

DEVELOPMENT SET

HELD-OUT DATA

► **Table 13.10** Data for parameter estimation exercise.

	docID	words in document	in $c = \text{China?}$
training set	1	Taipei Taiwan	yes
	2	Macao Taiwan Shanghai	yes
	3	Japan Sapporo	no
	4	Sapporo Osaka Taiwan	no
test set	5	Taiwan Taiwan Sapporo	?

tion about the test set was used in developing the classifier, the results of this experiment should be indicative of actual performance in practice.

This ideal often cannot be met; researchers tend to evaluate several systems on the same test set over a period of several years. But it is nevertheless highly important to not look at the test data and to run systems on it as sparingly as possible. Beginners often violate this rule, and their results lose validity because they have implicitly tuned their system to the test data simply by running many variant systems and keeping the tweaks to the system that worked best on the test set.



Exercise 13.6

[**]

Assume a situation where every document in the test collection has been assigned exactly one class, and that a classifier also assigns exactly one class to each document. This setup is called one-of classification (Section 14.5, page 306). Show that in one-of classification (i) the total number of false positive decisions equals the total number of false negative decisions and (ii) microaveraged F_1 and accuracy are identical.

Exercise 13.7

The class priors in Figure 13.2 are computed as the fraction of *documents* in the class as opposed to the fraction of *tokens* in the class. Why?

Exercise 13.8

The function `APPLYMULTINOMIALNB` in Figure 13.2 has time complexity $\Theta(L_a + |\mathcal{C}|L_a)$. How would you modify the function so that its time complexity is $\Theta(L_a + |\mathcal{C}|M_a)$?

Exercise 13.9

Based on the data in Table 13.10, (i) estimate a multinomial Naive Bayes classifier, (ii) apply the classifier to the test document, (iii) estimate a Bernoulli NB classifier, (iv) apply the classifier to the test document. You need not estimate parameters that you don't need for classifying the test document.

Exercise 13.10

Your task is to classify words as English or not English. Words are generated by a source with the following distribution:

event	word	English?	probability
1	ozb	no	4/9
2	uzu	no	4/9
3	zoo	yes	1/18
4	bun	yes	1/18

(i) Compute the parameters (priors and conditionals) of a multinomial NB classifier that uses the letters b, n, o, u, and z as features. Assume a training set that reflects the probability distribution of the source perfectly. Make the same independence assumptions that are usually made for a multinomial classifier that uses terms as features for text classification. Compute parameters using smoothing, in which computed-zero probabilities are smoothed into probability 0.01, and computed-nonzero probabilities are untouched. (This simplistic smoothing may cause $P(A) + P(\bar{A}) > 1$. Solutions are not required to correct this.) (ii) How does the classifier classify the word zoo? (iii) Classify the word zoo using a multinomial classifier as in part (i), but do not make the assumption of positional independence. That is, estimate separate parameters for each position in a word. You only need to compute the parameters you need for classifying zoo.

Exercise 13.11

What are the values of $I(U_t; C_c)$ and $X^2(\mathbb{D}, t, c)$ if term and class are completely independent? What are the values if they are completely dependent?

Exercise 13.12

The feature selection method in Equation (13.16) is most appropriate for the Bernoulli model. Why? How could one modify it for the multinomial model?

Exercise 13.13

INFORMATION GAIN

Features can also be selected according to *information gain* (IG), which is defined as:

$$IG(\mathbb{D}, t, c) = H(p_{\mathbb{D}}) - \sum_{x \in \{\mathbb{D}_{t+}, \mathbb{D}_{t-}\}} \frac{|x|}{|\mathbb{D}|} H(p_x)$$

where H is entropy, \mathbb{D} is the training set, and \mathbb{D}_{t+} , and \mathbb{D}_{t-} are the subset of \mathbb{D} with term t , and the subset of \mathbb{D} without term t , respectively. p_A is the class distribution in (sub)collection A , e.g., $p_A(c) = 0.25$, $p_A(\bar{c}) = 0.75$ if a quarter of the documents in A are in class c .

Show that mutual information and information gain are equivalent.

Exercise 13.14

Show that the two X^2 formulas (Equations (13.18) and (13.19)) are equivalent.

Exercise 13.15

In the χ^2 example on page 276 we have $|N_{11} - E_{11}| = |N_{10} - E_{10}| = |N_{01} - E_{01}| = |N_{00} - E_{00}|$. Show that this holds in general.

Exercise 13.16

χ^2 and mutual information do not distinguish between positively and negatively correlated features. Because most good text classification features are positively correlated (i.e., they occur more often in c than in \bar{c}), one may want to explicitly rule out the selection of negative indicators. How would you do this?

13.7 References and further reading

General introductions to statistical classification and machine learning can be found in (Hastie et al. 2001), (Mitchell 1997), and (Duda et al. 2000), including many important methods (e.g., decision trees and boosting) that we do not cover. A comprehensive review of text classification methods and results is (Sebastiani 2002). Manning and Schütze (1999, Chapter 16) give an accessible introduction to text classification with coverage of decision trees, perceptrons and maximum entropy models. More information on the superlinear time complexity of learning methods that are more accurate than Naive Bayes can be found in (Perkins et al. 2003) and (Joachims 2006a).

Maron and Kuhns (1960) described one of the first NB text classifiers. Lewis (1998) focuses on the history of NB classification. Bernoulli and multinomial models and their accuracy for different collections are discussed by McCalum and Nigam (1998). Eyheramendy et al. (2003) present additional NB models. Domingos and Pazzani (1997), Friedman (1997), and Hand and Yu (2001) analyze why NB performs well although its probability estimates are poor. The first paper also discusses NB's optimality when the independence assumptions are true of the data. Pavlov et al. (2004) propose a modified document representation that partially addresses the inappropriateness of the independence assumptions. Bennett (2000) attributes the tendency of NB probability estimates to be close to either 0 or 1 to the effect of document length. Ng and Jordan (2001) show that NB is sometimes (although rarely) superior to discriminative methods because it more quickly reaches its optimal error rate. The basic NB model presented in this chapter can be tuned for better effectiveness (Rennie et al. 2003; Kołcz and Yih 2007). The problem of concept drift and other reasons why state-of-the-art classifiers do not always excel in practice are discussed by Forman (2006) and Hand (2006).

Early uses of mutual information and χ^2 for feature selection in text classification are Lewis and Ringuette (1994) and Schütze et al. (1995), respectively. Yang and Pedersen (1997) review feature selection methods and their impact on classification effectiveness. They find that *pointwise mutual information* is not competitive with other methods. Yang and Pedersen refer to expected mutual information (Equation (13.16)) as information gain (see Exercise 13.13, page 285). (Snedecor and Cochran 1989) is a good reference for the χ^2 test in statistics, including the Yates' correction for continuity for 2×2 tables. Dunning (1993) discusses problems of the χ^2 test when counts are small. Nongreedy feature selection techniques are described by Hastie et al. (2001). Cohen (1995) discusses the pitfalls of using multiple significance tests and methods to avoid them. Forman (2004) evaluates different methods for feature selection for multiple classifiers.

David D. Lewis defines the ModApte split at www.daviddlewis.com/resources/testcollections/reuters21578 based on Apté et al. (1994). Lewis (1995) describes *utility measures* for the

evaluation of text classification systems. Yang and Liu (1999) employ significance tests in the evaluation of text classification methods.

Lewis et al. (2004) find that SVMs (Chapter 15) perform better on Reuters-RCV1 than kNN and Rocchio (Chapter 14).

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14 Vector space classification

CONTIGUITY
HYPOTHESIS

The document representation in Naive Bayes is a sequence of terms or a binary vector $\langle e_1, \dots, e_{|V|} \rangle \in \{0, 1\}^{|V|}$. In this chapter we adopt a different representation for text classification, the vector space model, developed in Chapter 6. It represents each document as a vector with one real-valued component, usually a tf-idf weight, for each term. Thus, the document space \mathbb{X} , the domain of the classification function γ , is $\mathbb{R}^{|V|}$. This chapter introduces a number of classification methods that operate on real-valued vectors.

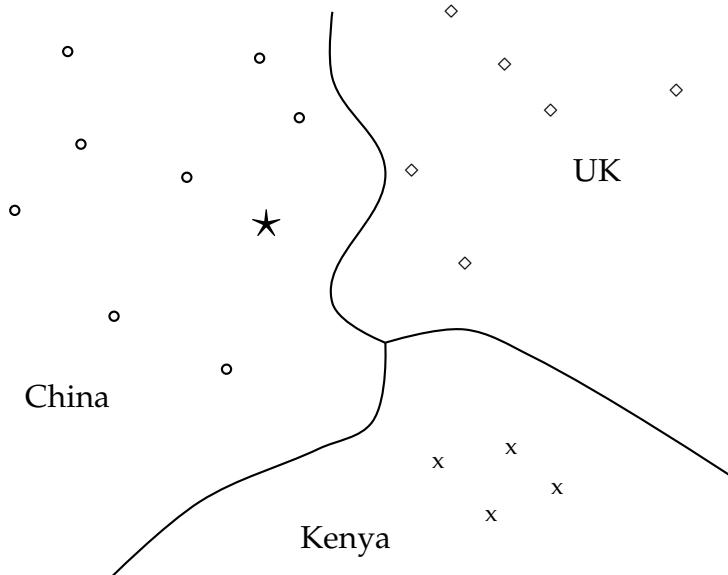
The basic hypothesis in using the vector space model for classification is the *contiguity hypothesis*.

Contiguity hypothesis. Documents in the same class form a contiguous region and regions of different classes do not overlap.

There are many classification tasks, in particular the type of text classification that we encountered in Chapter 13, where classes can be distinguished by word patterns. For example, documents in the class *China* tend to have high values on dimensions like Chinese, Beijing, and Mao whereas documents in the class *UK* tend to have high values for London, British and Queen. Documents of the two classes therefore form distinct contiguous regions as shown in Figure 14.1 and we can draw boundaries that separate them and classify new documents. How exactly this is done is the topic of this chapter.

Whether or not a set of documents is mapped into a contiguous region depends on the particular choices we make for the document representation: type of weighting, stop list etc. To see that the document representation is crucial, consider the two classes *written by a group* vs. *written by a single person*. Frequent occurrence of the first person pronoun I is evidence for the single-person class. But that information is likely deleted from the document representation if we use a stop list. If the document representation chosen is unfavorable, the contiguity hypothesis will not hold and successful vector space classification is not possible.

The same considerations that led us to prefer weighted representations, in particular length-normalized tf-idf representations, in Chapters 6 and 7 also



► **Figure 14.1** Vector space classification into three classes.

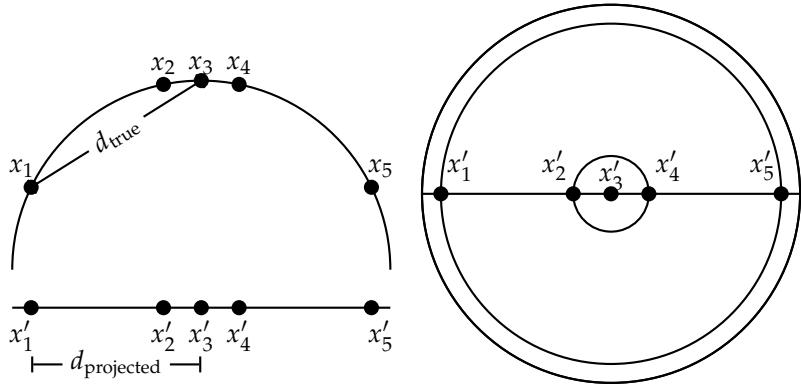
apply here. For example, a term with 5 occurrences in a document should get a higher weight than a term with one occurrence, but a weight 5 times larger would give too much emphasis to the term. Unweighted and unnormalized counts should not be used in vector space classification.

PROTOTYPE

We introduce two vector space classification methods in this chapter, Rocchio and kNN. Rocchio classification (Section 14.2) divides the vector space into regions centered on centroids or *prototypes*, one for each class, computed as the center of mass of all documents in the class. Rocchio classification is simple and efficient, but inaccurate if classes are not approximately spheres with similar radii.

kNN or k nearest neighbor classification (Section 14.3) assigns the majority class of the k nearest neighbors to a test document. kNN requires no explicit training and can use the unprocessed training set directly in classification. It is less efficient than other classification methods in classifying documents. If the training set is large, then kNN can handle non-spherical and other complex classes better than Rocchio.

A large number of text classifiers can be viewed as linear classifiers – classifiers that classify based on a simple linear combination of the features (Section 14.4). Such classifiers partition the space of features into regions separated by linear *decision hyperplanes*, in a manner to be detailed below. Because of the bias-variance tradeoff (Section 14.6) more complex nonlinear models



► **Figure 14.2** Projections of small areas of the unit sphere preserve distances. Left: A projection of the 2D semicircle to 1D. For the points x_1, x_2, x_3, x_4, x_5 at x coordinates $-0.9, -0.2, 0, 0.2, 0.9$ the distance $|x_2x_3| \approx 0.201$ only differs by 0.5% from $|x'_2x'_3| = 0.2$; but $|x_1x_3|/|x'_1x'_3| = d_{\text{true}}/d_{\text{projected}} \approx 1.06/0.9 \approx 1.18$ is an example of a large distortion (18%) when projecting a large area. Right: The corresponding projection of the 3D hemisphere to 2D.

are not systematically better than linear models. Nonlinear models have more parameters to fit on a limited amount of training data and are more likely to make mistakes for small and noisy data sets.

When applying two-class classifiers to problems with more than two classes, there are *one-of* tasks – a document must be assigned to exactly one of several mutually exclusive classes – and *any-of* tasks – a document can be assigned to any number of classes as we will explain in Section 14.5. Two-class classifiers solve any-of problems and can be combined to solve one-of problems.

14.1 Document representations and measures of relatedness in vector spaces

As in Chapter 6, we represent documents as vectors in $\mathbb{R}^{|V|}$ in this chapter. To illustrate properties of document vectors in vector classification, we will render these vectors as points in a plane as in the example in Figure 14.1. In reality, document vectors are length-normalized unit vectors that point to the surface of a hypersphere. We can view the 2D planes in our figures as projections onto a plane of the surface of a (hyper-)sphere as shown in Figure 14.2. Distances on the surface of the sphere and on the projection plane are approximately the same as long as we restrict ourselves to small areas of the surface and choose an appropriate projection (Exercise 14.1).

Decisions of many vector space classifiers are based on a notion of distance, e.g., when computing the nearest neighbors in kNN classification. We will use Euclidean distance in this chapter as the underlying distance measure. We observed earlier (Exercise 6.18, page 131) that there is a direct correspondence between cosine similarity and Euclidean distance for length-normalized vectors. In vector space classification, it rarely matters whether the relatedness of two documents is expressed in terms of similarity or distance.

However, in addition to documents, centroids or averages of vectors also play an important role in vector space classification. Centroids are not length-normalized. For unnormalized vectors, dot product, cosine similarity and Euclidean distance all have different behavior in general (Exercise 14.6). We will be mostly concerned with small local regions when computing the similarity between a document and a centroid, and the smaller the region the more similar the behavior of the three measures is.



Exercise 14.1

For small areas, distances on the surface of the hypersphere are approximated well by distances on its projection (Figure 14.2) because $\alpha \approx \sin \alpha$ for small angles. For what size angle is the distortion $\alpha / \sin(\alpha)$ (i) 1.01, (ii) 1.05 and (iii) 1.1?

14.2 Rocchio classification

DECISION BOUNDARY

ROCCHIO
CLASSIFICATION
CENTROID

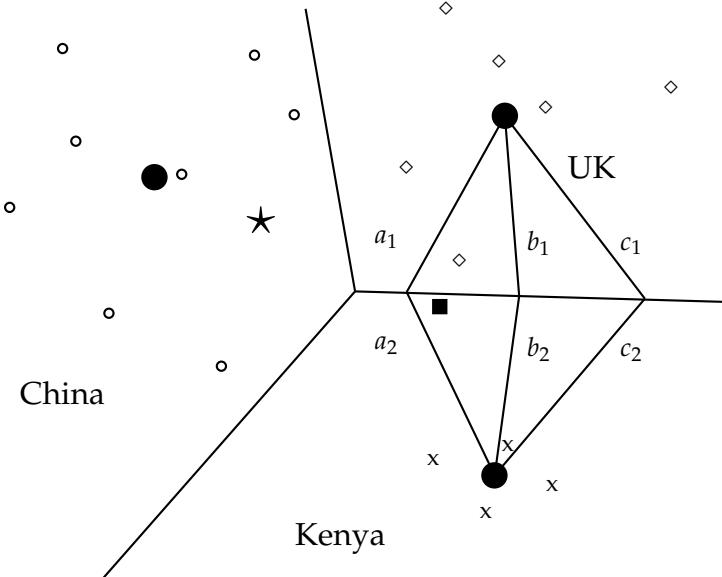
Figure 14.1 shows three classes, *China*, *UK* and *Kenya*, in a two-dimensional (2D) space. Documents are shown as circles, diamonds and X's. The boundaries in the figure, which we call *decision boundaries*, are chosen to separate the three classes, but are otherwise arbitrary. To classify a new document, depicted as a star in the figure, we determine the region it occurs in and assign it the class of that region – *China* in this case. Our task in vector space classification is to devise algorithms that compute good boundaries where “good” means high classification accuracy on data unseen during training.

Perhaps the best-known way of computing good class boundaries is *Rocchio classification*, which uses *centroids* to define the boundaries. The centroid of a class c is computed as the vector average or center of mass of its members:

$$(14.1) \quad \vec{\mu}(c) = \frac{1}{|D_c|} \sum_{d \in D_c} \vec{v}(d)$$

where D_c is the set of documents in \mathbb{D} whose class is c : $D_c = \{d : \langle d, c \rangle \in \mathbb{D}\}$. We denote the normalized vector of d by $\vec{v}(d)$ (Equation (6.11), page 122). Three example centroids are shown as solid circles in Figure 14.3.

The boundary between two classes in Rocchio classification is the set of points with equal distance from the two centroids. For example, $|a_1| = |a_2|$,



► **Figure 14.3** Rocchio classification.

$|b_1| = |b_2|$, and $|c_1| = |c_2|$ in the figure. This set of points is always a line. The generalization of a line in M -dimensional space is a hyperplane, which we define as the set of points \vec{x} that satisfy:

$$(14.2) \quad \vec{w}^T \vec{x} = b$$

NORMAL VECTOR

where \vec{w} is the M -dimensional *normal vector*¹ of the hyperplane and b is a constant. This definition of hyperplanes includes lines (any line in 2D can be defined by $w_1x_1 + w_2x_2 = b$) and 2-dimensional planes (any plane in 3D can be defined by $w_1x_1 + w_2x_2 + w_3x_3 = b$). A line divides a plane in two, a plane divides 3-dimensional space in two, and hyperplanes divide higher-dimensional spaces in two.

Thus, the boundaries of class regions in Rocchio classification are hyperplanes. The classification rule in Rocchio is to classify a point in accordance with the region it falls into. Equivalently, we determine the centroid $\vec{\mu}(c)$ that the point is closest to and then assign it to c . As an example, consider the star in Figure 14.3. It is located in the *China* region of the space and Rocchio therefore assigns it to *China*. We show the Rocchio algorithm in pseudocode in Figure 14.4.

1. Recall from basic linear algebra that $\vec{v} \cdot \vec{w} = \vec{v}^T \vec{w}$, i.e., the dot product of \vec{v} and \vec{w} equals the product by matrix multiplication of the transpose of \vec{v} and \vec{w} .

vector	term weights					
	Chinese	Japan	Tokyo	Macao	Beijing	Shanghai
\vec{d}_1	0	0	0	0	1.0	0
\vec{d}_2	0	0	0	0	0	1.0
\vec{d}_3	0	0	0	1.0	0	0
\vec{d}_4	0	0.71	0.71	0	0	0
\vec{d}_5	0	0.71	0.71	0	0	0
$\vec{\mu}_c$	0	0	0	0.33	0.33	0.33
$\vec{\mu}_{\bar{c}}$	0	0.71	0.71	0	0	0

► **Table 14.1** Vectors and class centroids for the data in Table 13.1.

 **Example 14.1:** Table 14.1 shows the tf-idf vector representations of the five documents in Table 13.1 (page 261), using the formula $(1 + \log_{10} \text{tf}_{t,d}) \log_{10}(4/\text{df}_t)$ if $\text{tf}_{t,d} > 0$ (Equation (6.14), page 127). The two class centroids are $\mu_c = 1/3 \cdot (\vec{d}_1 + \vec{d}_2 + \vec{d}_3)$ and $\mu_{\bar{c}} = 1/1 \cdot (\vec{d}_4)$. The distances of the test document from the centroids are $|\mu_c - \vec{d}_5| \approx 1.15$ and $|\mu_{\bar{c}} - \vec{d}_5| = 0.0$. Thus, Rocchio assigns d_5 to \bar{c} .

The separating hyperplane in this case has the following parameters:

$$\begin{aligned}\vec{w} &\approx (0 \ -0.71 \ -0.71 \ 1/3 \ 1/3 \ 1/3)^T \\ b &= -1/3\end{aligned}$$

See Exercise 14.15 for how to compute \vec{w} and b . We can easily verify that this hyperplane separates the documents as desired: $\vec{w}^T \vec{d}_1 \approx 0 \cdot 0 + -0.71 \cdot 0 + -0.71 \cdot 0 + 1/3 \cdot 0 + 1/3 \cdot 1.0 + 1/3 \cdot 0 = 1/3 > b$ (and, similarly, $\vec{w}^T \vec{d}_i > b$ for $i = 2$ and $i = 3$) and $\vec{w}^T \vec{d}_4 = -1 < b$. Thus, documents in c are above the hyperplane ($\vec{w}^T \vec{d} > b$) and documents in \bar{c} are below the hyperplane ($\vec{w}^T \vec{d} < b$).

The assignment criterion in Figure 14.4 is Euclidean distance (APPLYROCCHIO, line 1). An alternative is cosine similarity:

$$\text{Assign } d \text{ to class } c = \arg \max_{c'} \cos(\vec{\mu}(c'), \vec{v}(d))$$

As discussed in Section 14.1, the two assignment criteria will sometimes make different classification decisions. We present the Euclidean distance variant of Rocchio classification here because it emphasizes Rocchio's close correspondence to K-means clustering (Section 16.4, page 360).

Rocchio classification is a form of Rocchio relevance feedback (Section 9.1.1, page 178). The average of the relevant documents, corresponding to the most important component of the Rocchio vector in relevance feedback (Equation (9.3), page 182), is the centroid of the “class” of relevant documents. We omit the query component of the Rocchio formula in Rocchio classification since there is no query in text classification. Rocchio classification can be

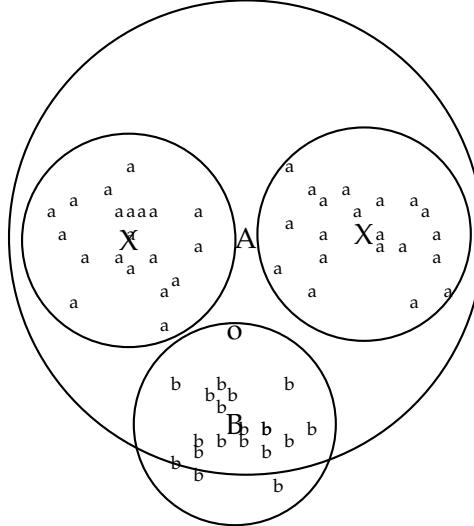
```

TRAINROCCHIO( $\mathbb{C}, \mathbb{D}$ )
1 for each  $c_j \in \mathbb{C}$ 
2 do  $D_j \leftarrow \{d : \langle d, c_j \rangle \in \mathbb{D}\}$ 
3  $\vec{\mu}_j \leftarrow \frac{1}{|D_j|} \sum_{d \in D_j} \vec{v}(d)$ 
4 return  $\{\vec{\mu}_1, \dots, \vec{\mu}_J\}$ 

APPLYROCCHIO( $\{\vec{\mu}_1, \dots, \vec{\mu}_J\}, d$ )
1 return  $\arg \min_j |\vec{\mu}_j - \vec{v}(d)|$ 

```

► **Figure 14.4** Rocchio classification: Training and testing.



► **Figure 14.5** The multimodal class “a” consists of two different clusters (small upper circles centered on X’s). Rocchio classification will misclassify “o” as “a” because it is closer to the centroid A of the “a” class than to the centroid B of the “b” class.

applied to $J > 2$ classes whereas Rocchio relevance feedback is designed to distinguish only two classes, relevant and nonrelevant.

In addition to respecting contiguity, the classes in Rocchio classification must be approximate spheres with similar radii. In Figure 14.3, the solid square just below the boundary between *UK* and *Kenya* is a better fit for the class *UK* since *UK* is more scattered than *Kenya*. But Rocchio assigns it to *Kenya* because it ignores details of the distribution of points in a class and only uses distance from the centroid for classification.

The assumption of sphericity also does not hold in Figure 14.5. We can-

mode	time complexity
training	$\Theta(\mathbb{D} L_{\text{ave}} + \mathcal{C} V)$
testing	$\Theta(L_a + \mathcal{C} M_a) = \Theta(\mathcal{C} M_a)$

► **Table 14.2** Training and test times for Rocchio classification. L_{ave} is the average number of tokens per document. L_a and M_a are the numbers of tokens and types, respectively, in the test document. Computing Euclidean distance between the class centroids and a document is $\Theta(|\mathcal{C}|M_a)$.

MULTIMODAL CLASS

not represent the “a” class well with a single prototype because it has two clusters. Rocchio often misclassifies this type of *multimodal class*. A text classification example for multimodality is a country like Burma, which changed its name to Myanmar in 1989. The two clusters before and after the name change need not be close to each other in space. We also encountered the problem of multimodality in relevance feedback (Section 9.1.2, page 184).

Two-class classification is another case where classes are rarely distributed like spheres with similar radii. Most two-class classifiers distinguish between a class like *China* that occupies a small region of the space and its widely scattered complement. Assuming equal radii will result in a large number of false positives. Most two-class classification problems therefore require a modified decision rule of the form:

$$\text{Assign } d \text{ to class } c \text{ iff } |\vec{\mu}(c) - \vec{v}(d)| < |\vec{\mu}(\bar{c}) - \vec{v}(d)| - b$$

for a positive constant b . As in Rocchio relevance feedback, the centroid of the negative documents is often not used at all, so that the decision criterion simplifies to $|\vec{\mu}(c) - \vec{v}(d)| < b'$ for a positive constant b' .

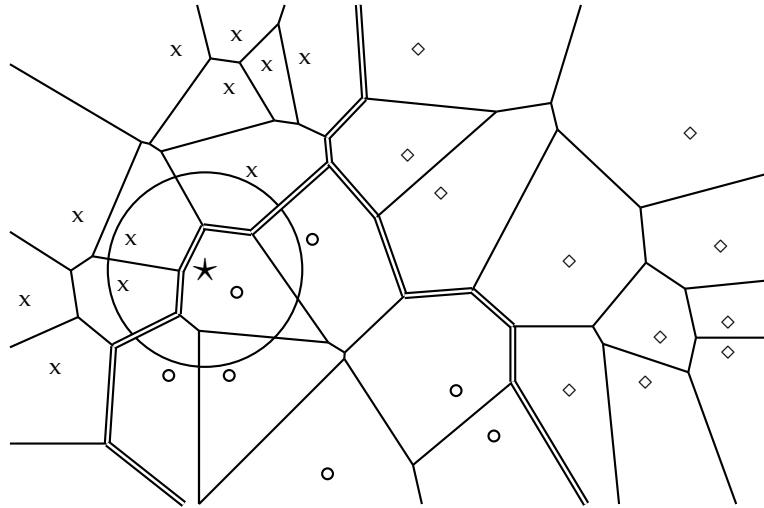
Table 14.2 gives the time complexity of Rocchio classification.² Adding all documents to their respective (unnormalized) centroid is $\Theta(|\mathbb{D}|L_{\text{ave}})$ (as opposed to $\Theta(|\mathbb{D}||V|)$) since we need only consider non-zero entries. Dividing each vector sum by the size of its class to compute the centroid is $\Theta(|V|)$. Overall, training time is linear in the size of the collection (cf. Exercise 13.1). Thus, Rocchio classification and Naive Bayes have the same linear training time complexity.

In the next section, we will introduce another vector space classification method, kNN, that deals better with classes that have non-spherical, disconnected or other irregular shapes.

**Exercise 14.2**[^{*}]

Show that Rocchio classification can assign a label to a document that is different from its training set label.

2. We write $\Theta(|\mathbb{D}|L_{\text{ave}})$ for $\Theta(T)$ and assume that the length of test documents is bounded as we did on page 262.



► **Figure 14.6** Voronoi tessellation and decision boundaries (double lines) in 1NN classification. The three classes are: X, circle and diamond.

14.3 *k* nearest neighbor

***k* NEAREST NEIGHBOR
CLASSIFICATION**

Unlike Rocchio, *k* nearest neighbor or *kNN* classification determines the decision boundary locally. For 1NN we assign each document to the class of its closest neighbor. For kNN we assign each document to the majority class of its k closest neighbors where k is a parameter. The rationale of kNN classification is that, based on the contiguity hypothesis, we expect a test document d to have the same label as the training documents located in the local region surrounding d .

**VORONOI
TESSELLATION**

Decision boundaries in 1NN are concatenated segments of the *Voronoi tessellation* as shown in Figure 14.6. The Voronoi tessellation of a set of objects decomposes space into Voronoi cells, where each object's cell consists of all points that are closer to the object than to other objects. In our case, the objects are documents. The Voronoi tessellation then partitions the plane into $|D|$ convex polygons, each containing its corresponding document (and no other) as shown in Figure 14.6, where a convex polygon is a convex region in 2-dimensional space bounded by lines.

For general $k \in \mathbb{N}$ in kNN, consider the region in the space for which the set of k nearest neighbors is the same. This again is a convex polygon and the space is partitioned into convex polygons, within each of which the set of k

```

TRAIN-KNN( $\mathbb{C}, \mathbb{D}$ )
1  $\mathbb{D}' \leftarrow \text{PREPROCESS}(\mathbb{D})$ 
2  $k \leftarrow \text{SELECT-K}(\mathbb{C}, \mathbb{D}')$ 
3 return  $\mathbb{D}', k$ 

APPLY-KNN( $\mathbb{C}, \mathbb{D}', k, d$ )
1  $S_k \leftarrow \text{COMPUTENEARESTNEIGHBORS}(\mathbb{D}', k, d)$ 
2 for each  $c_j \in \mathbb{C}$ 
3 do  $p_j \leftarrow |S_k \cap c_j|/k$ 
4 return  $\arg \max_j p_j$ 

```

► **Figure 14.7** kNN training (with preprocessing) and testing. p_j is an estimate for $P(c_j|S_k) = P(c_j|d)$. c_j denotes the set of all documents in the class c_j .

nearest neighbors is invariant (Exercise 14.11).³

1NN is not very robust. The classification decision of each test document relies on the class of a single training document, which may be incorrectly labeled or atypical. kNN for $k > 1$ is more robust. It assigns documents to the majority class of their k closest neighbors, with ties broken randomly.

There is a probabilistic version of this kNN classification algorithm. We can estimate the probability of membership in class c as the proportion of the k nearest neighbors in c . Figure 14.6 gives an example for $k = 3$. Probability estimates for class membership of the star are $\hat{P}(\text{circle class|star}) = 1/3$, $\hat{P}(\text{X class|star}) = 2/3$, and $\hat{P}(\text{diamond class|star}) = 0$. The 3nn estimate ($\hat{P}_1(\text{circle class|star}) = 1/3$) and the 1nn estimate ($\hat{P}_1(\text{circle class|star}) = 1$) differ with 3nn preferring the X class and 1nn preferring the circle class .

The parameter k in kNN is often chosen based on experience or knowledge about the classification problem at hand. It is desirable for k to be odd to make ties less likely. $k = 3$ and $k = 5$ are common choices, but much larger values between 50 and 100 are also used. An alternative way of setting the parameter is to select the k that gives best results on a held-out portion of the training set.

We can also weight the “votes” of the k nearest neighbors by their cosine

3. The generalization of a polygon to higher dimensions is a polytope. A polytope is a region in M -dimensional space bounded by $(M - 1)$ -dimensional hyperplanes. In M dimensions, the decision boundaries for kNN consist of segments of $(M - 1)$ -dimensional hyperplanes that form the Voronoi tessellation into convex polytopes for the training set of documents. The decision criterion of assigning a document to the majority class of its k nearest neighbors applies equally to $M = 2$ (tessellation into polygons) and $M > 2$ (tessellation into polytopes).

kNN with preprocessing of training set	
training	$\Theta(\mathcal{D} L_{ave})$
testing	$\Theta(L_a + \mathcal{D} M_{ave}M_a) = \Theta(\mathcal{D} M_{ave}M_a)$
kNN without preprocessing of training set	
training	$\Theta(1)$
testing	$\Theta(L_a + \mathcal{D} L_{ave}M_a) = \Theta(\mathcal{D} L_{ave}M_a)$

► **Table 14.3** Training and test times for kNN classification. M_{ave} is the average size of the vocabulary of documents in the collection.

similarity. In this scheme, a class's score is computed as:

$$\text{score}(c, d) = \sum_{d' \in S_k(d)} I_c(d') \cos(\vec{v}(d'), \vec{v}(d))$$

where $S_k(d)$ is the set of d 's k nearest neighbors and $I_c(d') = 1$ iff d' is in class c and 0 otherwise. We then assign the document to the class with the highest score. Weighting by similarities is often more accurate than simple voting. For example, if two classes have the same number of neighbors in the top k , the class with the more similar neighbors wins.

Figure 14.7 summarizes the kNN algorithm.



Example 14.2: The distances of the test document from the four training documents in Table 14.1 are $|\vec{d}_1 - \vec{d}_5| = |\vec{d}_2 - \vec{d}_5| = |\vec{d}_3 - \vec{d}_5| \approx 1.41$ and $|\vec{d}_4 - \vec{d}_5| = 0.0$. d_5 's nearest neighbor is therefore d_4 and 1NN assigns d_5 to d_4 's class, \bar{c} .



14.3.1 Time complexity and optimality of kNN

Table 14.3 gives the time complexity of kNN. kNN has properties that are quite different from most other classification algorithms. Training a kNN classifier simply consists of determining k and preprocessing documents. In fact, if we preselect a value for k and do not preprocess, then kNN requires no training at all. In practice, we have to perform preprocessing steps like tokenization. It makes more sense to preprocess training documents once as part of the training phase rather than repeatedly every time we classify a new test document.

Test time is $\Theta(|\mathcal{D}|M_{ave}M_a)$ for kNN. It is linear in the size of the training set as we need to compute the distance of each training document from the test document. Test time is independent of the number of classes J . kNN therefore has a potential advantage for problems with large J .

In kNN classification, we do not perform any estimation of parameters as we do in Rocchio classification (centroids) or in Naive Bayes (priors and conditional probabilities). kNN simply memorizes all examples in the training

MEMORY-BASED
LEARNING

set and then compares the test document to them. For this reason, kNN is also called *memory-based learning* or *instance-based learning*. It is usually desirable to have as much training data as possible in machine learning. But in kNN large training sets come with a severe efficiency penalty in classification.

Can kNN testing be made more efficient than $\Theta(|\mathcal{D}|M_{\text{ave}}M_a)$ or, ignoring the length of documents, more efficient than $\Theta(|\mathcal{D}|)$? There are fast kNN algorithms for small dimensionality M (Exercise 14.12). There are also approximations for large M that give error bounds for specific efficiency gains (see Section 14.7). These approximations have not been extensively tested for text classification applications, so it is not clear whether they can achieve much better efficiency than $\Theta(|\mathcal{D}|)$ without a significant loss of accuracy.

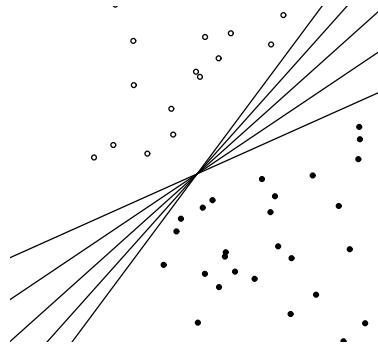
The reader may have noticed the similarity between the problem of finding nearest neighbors of a test document and ad hoc retrieval, where we search for the documents with the highest similarity to the query (Section 6.3.2, page 123). In fact, the two problems are both k nearest neighbor problems and only differ in the relative density of (the vector of) the test document in kNN (10s or 100s of non-zero entries) versus the sparseness of (the vector of) the query in ad hoc retrieval (usually fewer than 10 non-zero entries). We introduced the inverted index for efficient ad hoc retrieval in Section 1.1 (page 6). Is the inverted index also the solution for efficient kNN?

An inverted index restricts a search to those documents that have at least one term in common with the query. Thus in the context of kNN, the inverted index will be efficient if the test document has no term overlap with a large number of training documents. Whether this is the case depends on the classification problem. If documents are long and no stop list is used, then less time will be saved. But with short documents and a large stop list, an inverted index may well cut the average test time by a factor of 10 or more.

The search time in an inverted index is a function of the length of the postings lists of the terms in the query. Postings lists grow sublinearly with the length of the collection since the vocabulary increases according to Heaps' law – if the probability of occurrence of some terms increases, then the probability of occurrence of others must decrease. However, most new terms are infrequent. We therefore take the complexity of inverted index search to be $\Theta(T)$ (as discussed in Section 2.4.2, page 41) and, assuming average document length does not change over time, $\Theta(T) = \Theta(|\mathcal{D}|)$.

BAYES ERROR RATE

As we will see in the next chapter, kNN's effectiveness is close to that of the most accurate learning methods in text classification (Table 15.2, page 334). A measure of the quality of a learning method is its *Bayes error rate*, the average error rate of classifiers learned by it for a particular problem. kNN is not optimal for problems with a non-zero Bayes error rate – that is, for problems where even the best possible classifier has a non-zero classification error. The error of 1NN is asymptotically (as the training set increases) bounded by



► **Figure 14.8** There are an infinite number of hyperplanes that separate two linearly separable classes.

twice the Bayes error rate. That is, if the optimal classifier has an error rate of x , then 1NN has an asymptotic error rate of less than $2x$. This is due to the effect of noise – we already saw one example of noise in the form of noisy features in Section 13.5 (page 271), but noise can also take other forms as we will discuss in the next section. Noise affects two components of kNN: the test document and the closest training document. The two sources of noise are additive, so the overall error of 1NN is twice the optimal error rate. For problems with Bayes error rate 0, the error rate of 1NN will approach 0 as the size of the training set increases.



Exercise 14.3

Explain why kNN handles multimodal classes better than Rocchio.

14.4 Linear versus nonlinear classifiers

In this section, we show that the two learning methods Naive Bayes and Rocchio are instances of linear classifiers, the perhaps most important group of text classifiers, and contrast them with nonlinear classifiers. To simplify the discussion, we will only consider two-class classifiers in this section and define a *linear classifier* as a two-class classifier that decides class membership by comparing a linear combination of the features to a threshold.

In two dimensions, a linear classifier is a line. Five examples are shown in Figure 14.8. These lines have the functional form $w_1x_1 + w_2x_2 = b$. The classification rule of a linear classifier is to assign a document to c if $w_1x_1 + w_2x_2 > b$ and to \bar{c} if $w_1x_1 + w_2x_2 \leq b$. Here, $(x_1, x_2)^T$ is the two-dimensional vector representation of the document and $(w_1, w_2)^T$ is the parameter vector

```

APPLYLINEARCLASSIFIER( $\vec{w}, b, \vec{x}$ )
1    $score \leftarrow \sum_{i=1}^M w_i x_i$ 
2   if  $score > b$ 
3     then return 1
4   else return 0

```

► **Figure 14.9** Linear classification algorithm.

that defines (together with b) the decision boundary. An alternative geometric interpretation of a linear classifier is provided in Figure 15.7 (page 343).

We can generalize this 2D linear classifier to higher dimensions by defining a hyperplane as we did in Equation (14.2), repeated here as Equation (14.3):

$$(14.3) \quad \vec{w}^T \vec{x} = b$$

The assignment criterion then is: assign to c if $\vec{w}^T \vec{x} > b$ and to \bar{c} if $\vec{w}^T \vec{x} \leq b$. We call a hyperplane that we use as a linear classifier a *decision hyperplane*.

The corresponding algorithm for linear classification in M dimensions is shown in Figure 14.9. Linear classification at first seems trivial given the simplicity of this algorithm. However, the difficulty is in training the linear classifier, that is, in determining the parameters \vec{w} and b based on the training set. In general, some learning methods compute much better parameters than others where our criterion for evaluating the quality of a learning method is the effectiveness of the learned linear classifier on new data.

We now show that Rocchio and Naive Bayes are linear classifiers. To see this for Rocchio, observe that a vector \vec{x} is on the decision boundary if it has equal distance to the two class centroids:

$$(14.4) \quad |\vec{\mu}(c_1) - \vec{x}| = |\vec{\mu}(c_2) - \vec{x}|$$

Some basic arithmetic shows that this corresponds to a linear classifier with normal vector $\vec{w} = \vec{\mu}(c_1) - \vec{\mu}(c_2)$ and $b = 0.5 * (|\vec{\mu}(c_1)|^2 - |\vec{\mu}(c_2)|^2)$ (Exercise 14.15).

We can derive the linearity of Naive Bayes from its decision rule, which chooses the category c with the largest $\hat{P}(c|d)$ (Figure 13.2, page 260) where:

$$\hat{P}(c|d) \propto \hat{P}(c) \prod_{1 \leq k \leq n_d} \hat{P}(t_k|c)$$

and n_d is the number of tokens in the document that are part of the vocabulary. Denoting the complement category as \bar{c} , we obtain for the log odds:

$$(14.5) \quad \log \frac{\hat{P}(c|d)}{\hat{P}(\bar{c}|d)} = \log \frac{\hat{P}(c)}{\hat{P}(\bar{c})} + \sum_{1 \leq k \leq n_d} \log \frac{\hat{P}(t_k|c)}{\hat{P}(t_k|\bar{c})}$$

t_i	w_i	d_{1i}	d_{2i}	t_i	w_i	d_{1i}	d_{2i}
prime	0.70	0	1	dlrs	-0.71	1	1
rate	0.67	1	0	world	-0.35	1	0
interest	0.63	0	0	sees	-0.33	0	0
rates	0.60	0	0	year	-0.25	0	0
discount	0.46	1	0	group	-0.24	0	0
bundesbank	0.43	0	0	dlr	-0.24	0	0

► **Table 14.4** A linear classifier. The dimensions t_i and parameters w_i of a linear classifier for the class *interest* (as in interest rate) in Reuters-21578. The threshold is $b = 0$. Terms like *dlr* and *world* have negative weights because they are indicators for the competing class *currency*.

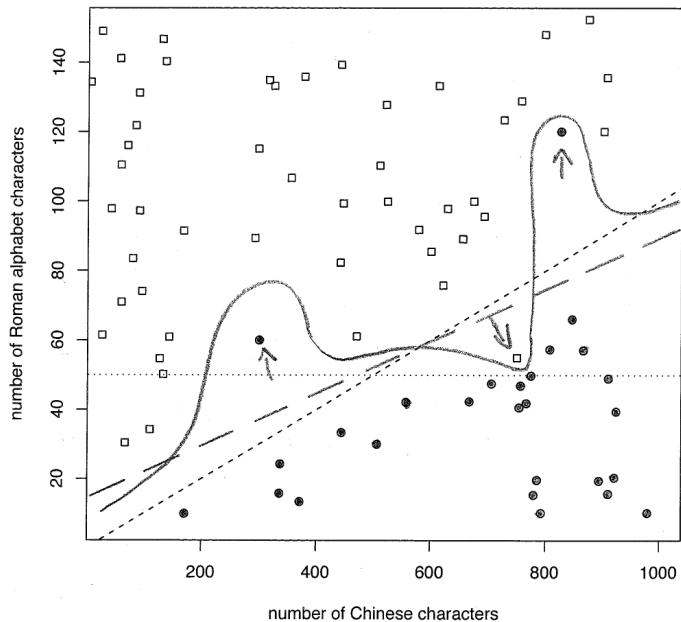
We choose class c if the odds are greater than 1 or, equivalently, if the log odds are greater than 0. It is easy to see that Equation (14.5) is an instance of Equation (14.3) for $w_i = \log[\hat{P}(t_i|c)/\hat{P}(t_i|\bar{c})]$, x_i = number of occurrences of t_i in d , and $b = -\log[\hat{P}(c)/\hat{P}(\bar{c})]$. Here, the index i , $1 \leq i \leq M$, refers to terms of the vocabulary (not to positions in d as k does; cf. Section 13.4.1, page 270) and \vec{x} and \vec{w} are M -dimensional vectors. So in log space, Naive Bayes is a linear classifier.

 **Example 14.3:** Table 14.4 defines a linear classifier for the category *interest* in Reuters-21578 (see Section 13.6, page 279). We assign document \vec{d}_1 “rate discount dtrs world” to *interest* since $\vec{w}^T \vec{d}_1 = 0.67 \cdot 1 + 0.46 \cdot 1 + (-0.71) \cdot 1 + (-0.35) \cdot 1 = 0.07 > 0 = b$. We assign \vec{d}_2 “prime dtrs” to the complement class (not in *interest*) since $\vec{w}^T \vec{d}_2 = -0.01 \leq b$. For simplicity, we assume a simple binary vector representation in this example: 1 for occurring terms, 0 for non-occurring terms.

CLASS BOUNDARY
NOISE DOCUMENT

Figure 14.10 is a graphical example of a *linear problem*, which we define to mean that the underlying distributions $P(d|c)$ and $P(d|\bar{c})$ of the two classes are separated by a line. We call this separating line the *class boundary*. It is the “true” boundary of the two classes and we distinguish it from the decision boundary that the learning method computes to approximate the class boundary.

As is typical in text classification, there are some *noise documents* in Figure 14.10 (marked with arrows) that do not fit well into the overall distribution of the classes. In Section 13.5 (page 271), we defined a noise feature as a misleading feature that, when included in the document representation, on average increases the classification error. Analogously, a noise document is a document that, when included in the training set, misleads the learning method and increases classification error. Intuitively, the underlying distribution partitions the representation space into areas with mostly ho-



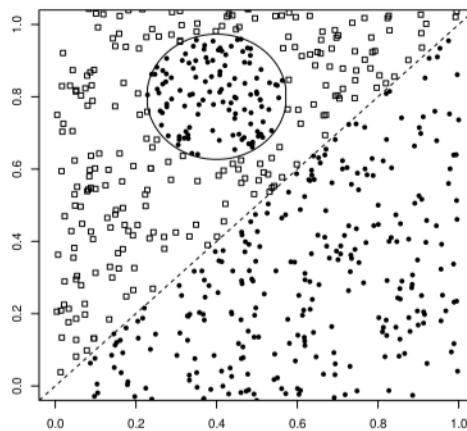
► **Figure 14.10** A linear problem with noise. In this hypothetical web page classification scenario, Chinese-only web pages are solid circles and mixed Chinese-English web pages are squares. The two classes are separated by a linear class boundary (dashed line, short dashes), except for three noise documents (marked with arrows).

mogeneous class assignments. A document that does not conform with the dominant class in its area is a noise document.

Noise documents are one reason why training a linear classifier is hard. If we pay too much attention to noise documents when choosing the decision hyperplane of the classifier, then it will be inaccurate on new data. More fundamentally, it is usually difficult to determine which documents are noise documents and therefore potentially misleading.

If there exists a hyperplane that perfectly separates the two classes, then we call the two classes *linearly separable*. In fact, if linear separability holds, then there is an infinite number of linear separators (Exercise 14.4) as illustrated by Figure 14.8, where the number of possible separating hyperplanes is infinite.

Figure 14.8 illustrates another challenge in training a linear classifier. If we are dealing with a linearly separable problem, then we need a criterion for selecting among all decision hyperplanes that perfectly separate the training data. In general, some of these hyperplanes will do well on new data, some



► **Figure 14.11** A nonlinear problem.

NONLINEAR
CLASSIFIER

will not.

An example of a *nonlinear classifier* is kNN. The nonlinearity of kNN is intuitively clear when looking at examples like Figure 14.6. The decision boundaries of kNN (the double lines in Figure 14.6) are locally linear segments, but in general have a complex shape that is not equivalent to a line in 2D or a hyperplane in higher dimensions.

Figure 14.11 is another example of a nonlinear problem: there is no good linear separator between the distributions $P(d|c)$ and $P(d|\bar{c})$ because of the circular “enclave” in the upper left part of the graph. Linear classifiers misclassify the enclave, whereas a nonlinear classifier like kNN will be highly accurate for this type of problem if the training set is large enough.

If a problem is nonlinear and its class boundaries cannot be approximated well with linear hyperplanes, then nonlinear classifiers are often more accurate than linear classifiers. If a problem is linear, it is best to use a simpler linear classifier.



Exercise 14.4

Prove that the number of linear separators of two classes is either infinite or zero.

14.5 Classification with more than two classes

We can extend two-class linear classifiers to $J > 2$ classes. The method to use depends on whether the classes are mutually exclusive or not.

ANY-OF
CLASSIFICATION

Classification for classes that are not mutually exclusive is called *any-of*, *multilabel*, or *multivalue classification*. In this case, a document can belong to several classes simultaneously, or to a single class, or to none of the classes. A decision on one class leaves all options open for the others. It is sometimes said that the classes are *independent* of each other, but this is misleading since the classes are rarely statistically independent in the sense defined on page 275. In terms of the formal definition of the classification problem in Equation (13.1) (page 256), we learn J different classifiers γ_j in any-of classification, each returning either c_j or \bar{c}_j : $\gamma_j(d) \in \{c_j, \bar{c}_j\}$.

Solving an any-of classification task with linear classifiers is straightforward:

1. Build a classifier for each class, where the training set consists of the set of documents in the class (positive labels) and its complement (negative labels).
2. Given the test document, apply each classifier separately. The decision of one classifier has no influence on the decisions of the other classifiers.

ONE-OF
CLASSIFICATION

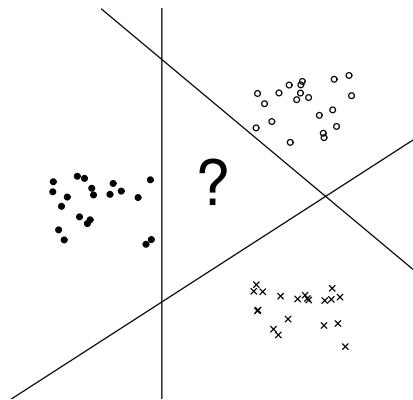
The second type of classification with more than two classes is *one-of classification*. Here, the classes are mutually exclusive. Each document must belong to exactly one of the classes. One-of classification is also called *multinomial*, *polytomous*⁴, *multiclass*, or *single-label classification*. Formally, there is a single classification function γ in one-of classification whose range is \mathbb{C} , i.e., $\gamma(d) \in \{c_1, \dots, c_J\}$. kNN is a (nonlinear) one-of classifier.

True one-of problems are less common in text classification than any-of problems. With classes like *UK*, *China*, *poultry*, or *coffee*, a document can be relevant to many topics simultaneously – as when the prime minister of the UK visits China to talk about the coffee and poultry trade.

Nevertheless, we will often make a one-of assumption, as we did in Figure 14.1, even if classes are not really mutually exclusive. For the classification problem of identifying the language of a document, the one-of assumption is a good approximation as most text is written in only one language. In such cases, imposing a one-of constraint can increase the classifier's effectiveness because errors that are due to the fact that the any-of classifiers assigned a document to either no class or more than one class are eliminated.

J hyperplanes do not divide $\mathbb{R}^{|V|}$ into J distinct regions as illustrated in Figure 14.12. Thus, we must use a combination method when using two-class linear classifiers for one-of classification. The simplest method is to

4. A synonym of polytomous is polychotomous.



► **Figure 14.12** J hyperplanes do not divide space into J disjoint regions.

rank classes and then select the top-ranked class. Geometrically, the ranking can be with respect to the distances from the J linear separators. Documents close to a class's separator are more likely to be misclassified, so the greater the distance from the separator, the more plausible it is that a positive classification decision is correct. Alternatively, we can use a direct measure of confidence to rank classes, e.g., probability of class membership. We can state this algorithm for one-of classification with linear classifiers as follows:

1. Build a classifier for each class, where the training set consists of the set of documents in the class (positive labels) and its complement (negative labels).
2. Given the test document, apply each classifier separately.
3. Assign the document to the class with
 - the maximum score,
 - the maximum confidence value,
 - or the maximum probability.

CONFUSION MATRIX

An important tool for analyzing the performance of a classifier for $J > 2$ classes is the *confusion matrix*. The confusion matrix shows for each pair of classes $\langle c_1, c_2 \rangle$, how many documents from c_1 were incorrectly assigned to c_2 . In Table 14.5, the classifier manages to distinguish the three financial classes *money-fx*, *trade*, and *interest* from the three agricultural classes *wheat*, *corn*, and *grain*, but makes many errors within these two groups. The confusion matrix can help pinpoint opportunities for improving the accuracy of the

true class	assigned class	<i>money-fx</i>	<i>trade</i>	<i>interest</i>	<i>wheat</i>	<i>corn</i>	<i>grain</i>
<i>money-fx</i>	95	0	10	0	0	0	0
<i>trade</i>	1	1	90	0	1	0	0
<i>interest</i>	13	0	0	0	0	0	0
<i>wheat</i>	0	0	1	34	3	7	0
<i>corn</i>	1	0	2	13	26	5	0
<i>grain</i>	0	0	2	14	5	10	0

► **Table 14.5** A confusion matrix for Reuters-21578. For example, 14 documents from *grain* were incorrectly assigned to *wheat*. Adapted from Picca et al. (2006).

system. For example, to address the second largest error in Table 14.5 (14 in the row *grain*), one could attempt to introduce features that distinguish *wheat* documents from *grain* documents.



Exercise 14.5

Create a training set of 300 documents, 100 each from three different languages (e.g., English, French, Spanish). Create a test set by the same procedure, but also add 100 documents from a fourth language. Train (i) a one-of classifier (ii) an any-of classifier on this training set and evaluate it on the test set. (iii) Are there any interesting differences in how the two classifiers behave on this task?



14.6 The bias-variance tradeoff

Nonlinear classifiers are more powerful than linear classifiers. For some problems, there exists a nonlinear classifier with zero classification error, but no such linear classifier. Does that mean that we should always use nonlinear classifiers for optimal effectiveness in statistical text classification?

To answer this question, we introduce the bias-variance tradeoff in this section, one of the most important concepts in machine learning. The tradeoff helps explain why there is no universally optimal learning method. Selecting an appropriate learning method is therefore an unavoidable part of solving a text classification problem.

Throughout this section, we use linear and nonlinear classifiers as prototypical examples of “less powerful” and “more powerful” learning, respectively. This is a simplification for a number of reasons. First, many nonlinear models subsume linear models as a special case. For instance, a nonlinear learning method like kNN will in some cases produce a linear classifier. Second, there are nonlinear models that are less complex than linear models. For instance, a quadratic polynomial with two parameters is less powerful than a 10,000-dimensional linear classifier. Third, the complexity of learning is not really a property of the classifier because there are many aspects

of learning (such as feature selection, cf. (Section 13.5, page 271), regularization, and constraints such as margin maximization in Chapter 15) that make a learning method either more powerful or less powerful without affecting the type of classifier that is the final result of learning – regardless of whether that classifier is linear or nonlinear. We refer the reader to the publications listed in Section 14.7 for a treatment of the bias-variance tradeoff that takes into account these complexities. In this section, linear and nonlinear classifiers will simply serve as proxies for weaker and stronger learning methods in text classification.

We first need to state our objective in text classification more precisely. In Section 13.1 (page 256), we said that we want to minimize classification error on the test set. The implicit assumption was that training documents and test documents are generated according to the same underlying distribution. We will denote this distribution $P(\langle d, c \rangle)$ where d is the document and c its label or class. Figures 13.4 and 13.5 were examples of generative models that decompose $P(\langle d, c \rangle)$ into the product of $P(c)$ and $P(d|c)$. Figures 14.10 and 14.11 depict generative models for $\langle d, c \rangle$ with $d \in \mathbb{R}^2$ and $c \in \{\text{square, solid circle}\}$.

In this section, instead of using the number of correctly classified test documents (or, equivalently, the error rate on test documents) as evaluation measure, we adopt an evaluation measure that addresses the inherent uncertainty of labeling. In many text classification problems, a given document representation can arise from documents belonging to different classes. This is because documents from different classes can be mapped to the same document representation. For example, the one-sentence documents *China sues France* and *France sues China* are mapped to the same document representation $d' = \{\text{China, France, sues}\}$ in a bag of words model. But only the latter document is relevant to the class $c' = \text{legal actions brought by France}$ (which might be defined, for example, as a standing query by an international trade lawyer).

To simplify the calculations in this section, we do not count the number of errors on the test set when evaluating a classifier, but instead look at how well the classifier estimates the conditional probability $P(c|d)$ of a document being in a class. In the above example, we might have $P(c'|d') = 0.5$.

Our goal in text classification then is to find a classifier γ such that, averaged over documents d , $\gamma(d)$ is as close as possible to the true probability $P(c|d)$. We measure this using mean squared error:

$$(14.6) \quad \text{MSE}(\gamma) = E_d[\gamma(d) - P(c|d)]^2$$

where E_d is the expectation with respect to $P(d)$. The mean squared error term gives partial credit for decisions by γ that are close if not completely right.

$$\begin{aligned}
 (14.8) \quad E[x - \alpha]^2 &= Ex^2 - 2Ex\alpha + \alpha^2 \\
 &= (Ex)^2 - 2Ex\alpha + \alpha^2 \\
 &\quad + Ex^2 - 2(Ex)^2 + (Ex)^2 \\
 &= [Ex - \alpha]^2 \\
 &\quad + Ex^2 - E2x(Ex) + E(Ex)^2 \\
 &= [Ex - \alpha]^2 + E[x - Ex]^2
 \end{aligned}$$

$$\begin{aligned}
 (14.9) \quad E_{\mathbb{D}}E_d[\Gamma_{\mathbb{D}}(d) - P(c|d)]^2 &= E_dE_{\mathbb{D}}[\Gamma_{\mathbb{D}}(d) - P(c|d)]^2 \\
 &= E_d[[E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d) - P(c|d)]^2 \\
 &\quad + E_{\mathbb{D}}[\Gamma_{\mathbb{D}}(d) - E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)]^2]
 \end{aligned}$$

► **Figure 14.13** Arithmetic transformations for the bias-variance decomposition. For the derivation of Equation (14.9), we set $\alpha = P(c|d)$ and $x = \Gamma_{\mathbb{D}}(d)$ in Equation (14.8).

OPTIMAL CLASSIFIER

We define a classifier γ to be *optimal* for a distribution $P(\langle d, c \rangle)$ if it minimizes $MSE(\gamma)$.

Minimizing MSE is a desideratum for *classifiers*. We also need a criterion for *learning methods*. Recall that we defined a learning method Γ as a function that takes a labeled training set \mathbb{D} as input and returns a classifier γ .

LEARNING ERROR

For learning methods, we adopt as our goal to find a Γ that, averaged over training sets, learns classifiers γ with minimal MSE. We can formalize this as minimizing *learning error*:

$$(14.7) \quad \text{learning-error}(\Gamma) = E_{\mathbb{D}}[\text{MSE}(\Gamma(\mathbb{D}))]$$

where $E_{\mathbb{D}}$ is the expectation over labeled training sets. To keep things simple, we can assume that training sets have a fixed size – the distribution $P(\langle d, c \rangle)$ then defines a distribution $P(\mathbb{D})$ over training sets.

OPTIMAL LEARNING METHOD

We can use learning error as a criterion for selecting a learning method in statistical text classification. A learning method Γ is *optimal* for a distribution $P(\mathbb{D})$ if it minimizes the learning error.

Writing $\Gamma_{\mathbb{D}}$ for $\Gamma(\mathbb{D})$ for better readability, we can transform Equation (14.7) as follows:

$$\begin{aligned}
 (14.10) \quad \text{learning-error}(\Gamma) &= E_{\mathbb{D}}[\text{MSE}(\Gamma_{\mathbb{D}})] \\
 (14.11) &= E_{\mathbb{D}}E_d[\Gamma_{\mathbb{D}}(d) - P(c|d)]^2 \\
 &= E_d[\text{bias}(\Gamma, d) + \text{variance}(\Gamma, d)]
 \end{aligned}$$

$$(14.12) \quad \text{bias}(\Gamma, d) = [P(c|d) - E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)]^2$$

$$(14.13) \quad \text{variance}(\Gamma, d) = E_{\mathbb{D}}[\Gamma_{\mathbb{D}}(d) - E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)]^2$$

where the equivalence between Equations (14.10) and (14.11) is shown in Equation (14.9) in Figure 14.13. Note that d and \mathbb{D} are independent of each other. In general, for a random document d and a random training set \mathbb{D} , \mathbb{D} does not contain a labeled instance of d .

BIAS

Bias is the squared difference between $P(c|d)$, the true conditional probability of d being in c , and $\Gamma_{\mathbb{D}}(d)$, the prediction of the learned classifier, averaged over training sets. Bias is large if the learning method produces classifiers that are consistently wrong. Bias is small if (i) the classifiers are consistently right or (ii) different training sets cause errors on different documents or (iii) different training sets cause positive and negative errors on the same documents, but that average out to close to 0. If one of these three conditions holds, then $E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)$, the expectation over all training sets, is close to $P(c|d)$.

Linear methods like Rocchio and Naive Bayes have a high bias for non-linear problems because they can only model one type of class boundary, a linear hyperplane. If the generative model $P(\langle d, c \rangle)$ has a complex nonlinear class boundary, the bias term in Equation (14.11) will be high because a large number of points will be consistently misclassified. For example, the circular enclave in Figure 14.11 does not fit a linear model and will be misclassified consistently by linear classifiers.

We can think of bias as resulting from our domain knowledge (or lack thereof) that we build into the classifier. If we know that the true boundary between the two classes is linear, then a learning method that produces linear classifiers is more likely to succeed than a nonlinear method. But if the true class boundary is not linear and we incorrectly bias the classifier to be linear, then classification accuracy will be low on average.

Nonlinear methods like kNN have low bias. We can see in Figure 14.6 that the decision boundaries of kNN are variable – depending on the distribution of documents in the training set, learned decision boundaries can vary greatly. As a result, each document has a chance of being classified correctly for some training sets. The average prediction $E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)$ is therefore closer to $P(c|d)$ and bias is smaller than for a linear learning method.

VARIANCE

Variance is the variation of the prediction of learned classifiers: the average squared difference between $\Gamma_{\mathbb{D}}(d)$ and its average $E_{\mathbb{D}}\Gamma_{\mathbb{D}}(d)$. Variance is large if different training sets \mathbb{D} give rise to very different classifiers $\Gamma_{\mathbb{D}}$. It is small if the training set has a minor effect on the classification decisions $\Gamma_{\mathbb{D}}$ makes, be they correct or incorrect. Variance measures how inconsistent the decisions are, not whether they are correct or incorrect.

Linear learning methods have low variance because most randomly drawn training sets produce similar decision hyperplanes. The decision lines pro-

duced by linear learning methods in Figures 14.10 and 14.11 will deviate slightly from the main class boundaries, depending on the training set, but the class assignment for the vast majority of documents (with the exception of those close to the main boundary) will not be affected. The circular enclave in Figure 14.11 will be consistently misclassified.

Nonlinear methods like kNN have high variance. It is apparent from Figure 14.6 that kNN can model very complex boundaries between two classes. It is therefore sensitive to noise documents of the sort depicted in Figure 14.10. As a result the variance term in Equation (14.11) is large for kNN: Test documents are sometimes misclassified – if they happen to be close to a noise document in the training set – and sometimes correctly classified – if there are no noise documents in the training set near them. This results in high variation from training set to training set.

OVERFITTING

High-variance learning methods are prone to *overfitting* the training data. The goal in classification is to fit the training data to the extent that we capture true properties of the underlying distribution $P(\langle d, c \rangle)$. In overfitting, the learning method also learns from noise. Overfitting increases MSE and frequently is a problem for high-variance learning methods.

MEMORY CAPACITY

We can also think of variance as the *model complexity* or, equivalently, *memory capacity* of the learning method – how detailed a characterization of the training set it can remember and then apply to new data. This capacity corresponds to the number of independent parameters available to fit the training set. Each kNN neighborhood S_k makes an independent classification decision. The parameter in this case is the estimate $\hat{P}(c|S_k)$ from Figure 14.7. Thus, kNN's capacity is only limited by the size of the training set. It can memorize arbitrarily large training sets. In contrast, the number of parameters of Rocchio is fixed – J parameters per dimension, one for each centroid – and independent of the size of the training set. The Rocchio classifier (in form of the centroids defining it) cannot “remember” fine-grained details of the distribution of the documents in the training set.

BIAS-VARIANCE TRADEOFF

According to Equation (14.7), our goal in selecting a learning method is to minimize learning error. The fundamental insight captured by Equation (14.11), which we can succinctly state as: learning-error = bias + variance, is that the learning error has two components, bias and variance, which in general cannot be minimized simultaneously. When comparing two learning methods Γ_1 and Γ_2 , in most cases the comparison comes down to one method having higher bias and lower variance and the other lower bias and higher variance. The decision for one learning method vs. another is then not simply a matter of selecting the one that reliably produces good classifiers across training sets (small variance) or the one that can learn classification problems with very difficult decision boundaries (small bias). Instead, we have to weigh the respective merits of bias and variance in our application and choose accordingly. This tradeoff is called the *bias-variance tradeoff*.

Figure 14.10 provides an illustration, which is somewhat contrived, but will be useful as an example for the tradeoff. Some Chinese text contains English words written in the Roman alphabet like CPU, ONLINE, and GPS. Consider the task of distinguishing Chinese-only web pages from mixed Chinese-English web pages. A search engine might offer Chinese users without knowledge of English (but who understand loanwords like CPU) the option of filtering out mixed pages. We use two features for this classification task: number of Roman alphabet characters and number of Chinese characters on the web page. As stated earlier, the distribution $P(\langle d, c \rangle)$ of the generative model generates most mixed (respectively, Chinese) documents above (respectively, below) the short-dashed line, but there are a few noise documents.

In Figure 14.10, we see three classifiers:

- **One-feature classifier.** Shown as a dotted horizontal line. This classifier uses only one feature, the number of Roman alphabet characters. Assuming a learning method that minimizes the number of misclassifications in the training set, the position of the horizontal decision boundary is not greatly affected by differences in the training set (e.g., noise documents). So a learning method producing this type of classifier has low variance. But its bias is high since it will consistently misclassify squares in the lower left corner and “solid circle” documents with more than 50 Roman characters.
- **Linear classifier.** Shown as a dashed line with long dashes. Learning linear classifiers has less bias since only noise documents and possibly a few documents close to the boundary between the two classes are misclassified. The variance is higher than for the one-feature classifiers, but still small: The dashed line with long dashes deviates only slightly from the true boundary between the two classes, and so will almost all linear decision boundaries learned from training sets. Thus, very few documents (documents close to the class boundary) will be inconsistently classified.
- **“Fit-training-set-perfectly” classifier.** Shown as a solid line. Here, the learning method constructs a decision boundary that perfectly separates the classes in the training set. This method has the lowest bias because there is no document that is consistently misclassified – the classifiers sometimes even get noise documents in the test set right. But the variance of this learning method is high. Because noise documents can move the decision boundary arbitrarily, test documents close to noise documents in the training set will be misclassified – something that a linear learning method is unlikely to do.

It is perhaps surprising that so many of the best-known text classification algorithms are linear. Some of these methods, in particular linear SVMs, reg-

ularized logistic regression and regularized linear regression, are among the most effective known methods. The bias-variance tradeoff provides insight into their success. Typical classes in text classification are complex and seem unlikely to be modeled well linearly. However, this intuition is misleading for the high-dimensional spaces that we typically encounter in text applications. With increased dimensionality, the likelihood of linear separability increases rapidly (Exercise 14.17). Thus, linear models in high-dimensional spaces are quite powerful despite their linearity. Even more powerful nonlinear learning methods can model decision boundaries that are more complex than a hyperplane, but they are also more sensitive to noise in the training data. Nonlinear learning methods sometimes perform better if the training set is large, but by no means in all cases.

14.7 References and further reading

ROUTING
FILTERING

PUSH MODEL
PULL MODEL

CENTROID-BASED
CLASSIFICATION

As discussed in Chapter 9, Rocchio relevance feedback is due to Rocchio (1971). Joachims (1997) presents a probabilistic analysis of the method. Rocchio classification was widely used as a classification method in TREC in the 1990s (Buckley et al. 1994a;b, Voorhees and Harman 2005). Initially, it was used as a form of *routing*. Routing merely ranks documents according to relevance to a class without assigning them. Early work on *filtering*, a true classification approach that makes an assignment decision on each document, was published by Ittner et al. (1995) and Schapire et al. (1998). The definition of routing we use here should not be confused with another sense. Routing can also refer to the electronic distribution of documents to subscribers, the so-called *push model* of document distribution. In a *pull model*, each transfer of a document to the user is initiated by the user – for example, by means of search or by selecting it from a list of documents on a news aggregation website.

Some authors restrict the name *Rocchio classification* to two-class problems and use the terms *cluster-based* (Iwayama and Tokunaga 1995) and *centroid-based classification* (Han and Karypis 2000, Tan and Cheng 2007) for Rocchio classification with $J > 2$.

A more detailed treatment of kNN can be found in (Hastie et al. 2001), including methods for tuning the parameter k . An example of an approximate fast kNN algorithm is locality-based hashing (Andoni et al. 2006). Kleinberg (1997) presents an approximate $\Theta((M \log^2 M)(M + \log N))$ kNN algorithm (where M is the dimensionality of the space and N the number of data points), but at the cost of exponential storage requirements: $\Theta((N \log M)^{2M})$. Indyk (2004) surveys nearest neighbor methods in high-dimensional spaces. Early work on kNN in text classification was motivated by the availability of massively parallel hardware architectures (Creecy et al. 1992). Yang (1994)

uses an inverted index to speed up kNN classification. The optimality result for 1NN (twice the Bayes error rate asymptotically) is due to [Cover and Hart \(1967\)](#).

The effectiveness of Rocchio classification and kNN is highly dependent on careful parameter tuning (in particular, the parameters b' for Rocchio on page 296 and k for kNN), feature engineering (Section 15.3, page 334) and feature selection (Section 13.5, page 271). [Buckley and Salton \(1995\)](#), [Schapire et al. \(1998\)](#), [Yang and Kisiel \(2003\)](#) and [Moschitti \(2003\)](#) address these issues for Rocchio and [Yang \(2001\)](#) and [Ault and Yang \(2002\)](#) for kNN. [Zavrel et al. \(2000\)](#) compare feature selection methods for kNN.

The bias-variance tradeoff was introduced by [Geman et al. \(1992\)](#). The derivation in Section 14.6 is for $\text{MSE}(\gamma)$, but the tradeoff applies to many loss functions (cf. [Friedman \(1997\)](#), [Domingos \(2000\)](#)). [Schütze et al. \(1995\)](#) and [Lewis et al. \(1996\)](#) discuss linear classifiers for text and [Hastie et al. \(2001\)](#) linear classifiers in general. Readers interested in the algorithms mentioned, but not described in this chapter may wish to consult [Bishop \(2006\)](#) for neural networks, [Hastie et al. \(2001\)](#) for linear and logistic regression, and [Minsky and Papert \(1988\)](#) for the perceptron algorithm. [Anagnostopoulos et al. \(2006\)](#) show that an inverted index can be used for highly efficient document classification with any linear classifier, provided that the classifier is still effective when trained on a modest number of features via feature selection.

We have only presented the simplest method for combining two-class classifiers into a one-of classifier. Another important method is the use of error-correcting codes, where a vector of decisions of different two-class classifiers is constructed for each document. A test document's decision vector is then “corrected” based on the distribution of decision vectors in the training set, a procedure that incorporates information from all two-class classifiers and their correlations into the final classification decision ([Dietterich and Bakiri 1995](#)). [Ghamrawi and McCallum \(2005\)](#) also exploit dependencies between classes in any-of classification. [Allwein et al. \(2000\)](#) propose a general framework for combining two-class classifiers.

14.8 Exercises

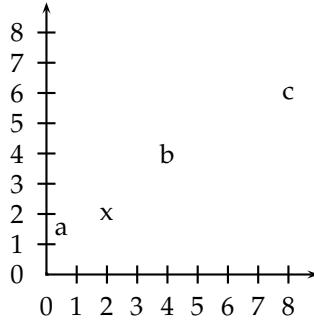


Exercise 14.6

In Figure 14.14, which of the three vectors \vec{a} , \vec{b} , and \vec{c} is (i) most similar to \vec{x} according to dot product similarity, (ii) most similar to \vec{x} according to cosine similarity, (iii) closest to \vec{x} according to Euclidean distance?

Exercise 14.7

Download Reuters-21578 and train and test Rocchio and kNN classifiers for the classes *acquisitions*, *corn*, *crude*, *earn*, *grain*, *interest*, *money-fx*, *ship*, *trade*, and *wheat*. Use the ModApte split. You may want to use one of a number of software packages that im-



► **Figure 14.14** Example for differences between Euclidean distance, dot product similarity and cosine similarity. The vectors are $\vec{a} = (0.5 \ 1.5)^T$, $\vec{x} = (2 \ 2)^T$, $\vec{b} = (4 \ 4)^T$, and $\vec{c} = (8 \ 6)^T$.

plement Rocchio classification and kNN classification, for example, the Bow toolkit (McCallum 1996).

Exercise 14.8

Download 20 Newsgroups (page 154) and train and test Rocchio and kNN classifiers for its 20 classes.

Exercise 14.9

Show that the decision boundaries in Rocchio classification are, as in kNN, given by the Voronoi tessellation.

Exercise 14.10

[*]

Computing the distance between a dense centroid and a sparse vector is $\Theta(M)$ for a naive implementation that iterates over all M dimensions. Based on the equality $\sum(x_i - \mu_i)^2 = 1.0 + \sum\mu_i^2 - 2\sum x_i \mu_i$ and assuming that $\sum\mu_i^2$ has been precomputed, write down an algorithm that is $\Theta(M_a)$ instead, where M_a is the number of distinct terms in the test document.

Exercise 14.11

[***]

Prove that the region of the plane consisting of all points with the same k nearest neighbors is a convex polygon.

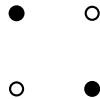
Exercise 14.12

Design an algorithm that performs an efficient 1NN search in 1 dimension (where efficiency is with respect to the number of documents N). What is the time complexity of the algorithm?

Exercise 14.13

[***]

Design an algorithm that performs an efficient 1NN search in 2 dimensions with at most polynomial (in N) preprocessing time.



► **Figure 14.15** A simple non-separable set of points.

Exercise 14.14

[***]

Can one design an exact efficient algorithm for 1NN for very large M along the ideas you used to solve the last exercise?

Exercise 14.15

Show that Equation (14.4) defines a hyperplane with $\vec{w} = \vec{\mu}(c_1) - \vec{\mu}(c_2)$ and $b = 0.5 * (|\vec{\mu}(c_1)|^2 - |\vec{\mu}(c_2)|^2)$.

Exercise 14.16

We can easily construct non-separable data sets in high dimensions by embedding a non-separable set like the one shown in Figure 14.15. Consider embedding Figure 14.15 in 3D and then perturbing the 4 points slightly (i.e., moving them a small distance in a random direction). Why would you expect the resulting configuration to be linearly separable? How likely is then a non-separable set of $m \ll M$ points in M -dimensional space?

Exercise 14.17

Assuming two classes, show that the percentage of non-separable assignments of the vertices of a hypercube decreases with dimensionality M for $M > 1$. For example, for $M = 1$ the proportion of non-separable assignments is 0, for $M = 2$, it is 2/16. One of the two non-separable cases for $M = 2$ is shown in Figure 14.15, the other is its mirror image. Solve the exercise either analytically or by simulation.

Exercise 14.18

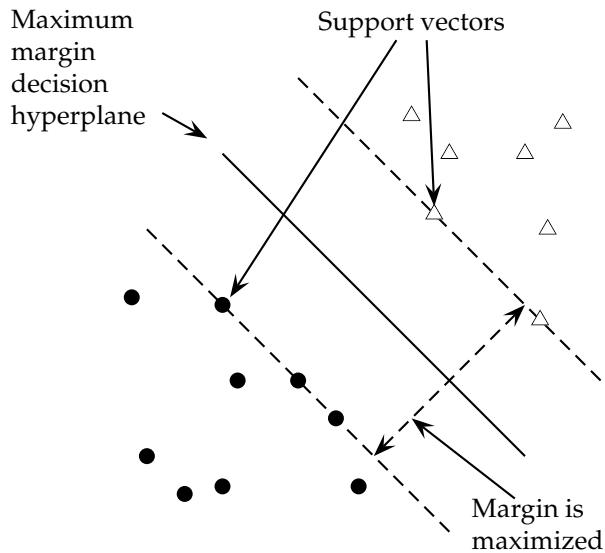
Although we point out the similarities of Naive Bayes with linear vector space classifiers, it does not make sense to represent count vectors (the document representations in NB) in a continuous vector space. There is however a formalization of NB that is analogous to Rocchio. Show that NB assigns a document to the class (represented as a parameter vector) whose Kullback-Leibler (KL) divergence (Section 12.4, page 251) to the document (represented as a count vector as in Section 13.4.1 (page 270), normalized to sum to 1) is smallest.

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15 *Support vector machines and machine learning on documents*

Improving classifier effectiveness has been an area of intensive machine-learning research over the last two decades, and this work has led to a new generation of state-of-the-art classifiers, such as support vector machines, boosted decision trees, regularized logistic regression, neural networks, and random forests. Many of these methods, including support vector machines (SVMs), the main topic of this chapter, have been applied with success to information retrieval problems, particularly text classification. An SVM is a kind of large-margin classifier: it is a vector space based machine learning method where the goal is to find a decision boundary between two classes that is maximally far from any point in the training data (possibly discounting some points as outliers or noise).

We will initially motivate and develop SVMs for the case of two-class data sets that are separable by a linear classifier (Section 15.1), and then extend the model in Section 15.2 to non-separable data, multi-class problems, and non-linear models, and also present some additional discussion of SVM performance. The chapter then moves to consider the practical deployment of text classifiers in Section 15.3: what sorts of classifiers are appropriate when, and how can you exploit domain-specific text features in classification? Finally, we will consider how the machine learning technology that we have been building for text classification can be applied back to the problem of learning how to rank documents in ad hoc retrieval (Section 15.4). While several machine learning methods have been applied to this task, use of SVMs has been prominent. Support vector machines are not necessarily better than other machine learning methods (except perhaps in situations with little training data), but they perform at the state-of-the-art level and have much current theoretical and empirical appeal.

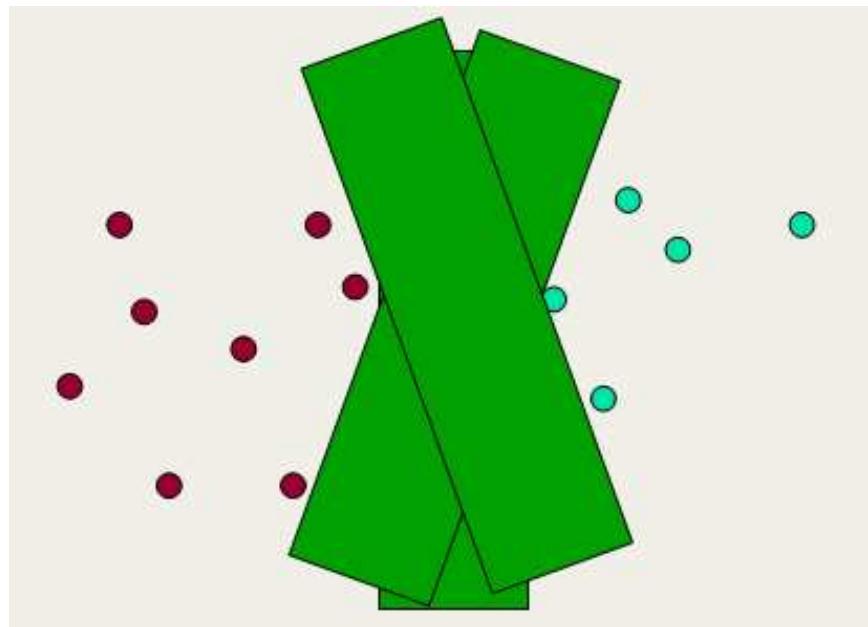


► **Figure 15.1** The support vectors are the 5 points right up against the margin of the classifier.

15.1 Support vector machines: The linearly separable case

MARGIN
SUPPORT VECTOR

For two-class, separable training data sets, such as the one in Figure 14.8 (page 301), there are lots of possible linear separators. Intuitively, a decision boundary drawn in the middle of the void between data items of the two classes seems better than one which approaches very close to examples of one or both classes. While some learning methods such as the perceptron algorithm (see references in Section 14.7, page 314) find just any linear separator, others, like Naive Bayes, search for the best linear separator according to some criterion. The SVM in particular defines the criterion to be looking for a decision surface that is maximally far away from any data point. This distance from the decision surface to the closest data point determines the *margin* of the classifier. This method of construction necessarily means that the decision function for an SVM is fully specified by a (usually small) subset of the data which defines the position of the separator. These points are referred to as the *support vectors* (in a vector space, a point can be thought of as a vector between the origin and that point). Figure 15.1 shows the margin and support vectors for a sample problem. Other data points play no part in determining the decision surface that is chosen.



► **Figure 15.2** An intuition for large-margin classification. Insisting on a large margin reduces the capacity of the model: the range of angles at which the fat decision surface can be placed is smaller than for a decision hyperplane (cf. Figure 14.8, page 301).

Maximizing the margin seems good because points near the decision surface represent very uncertain classification decisions: there is almost a 50% chance of the classifier deciding either way. A classifier with a large margin makes no low certainty classification decisions. This gives you a classification safety margin: a slight error in measurement or a slight document variation will not cause a misclassification. Another intuition motivating SVMs is shown in Figure 15.2. By construction, an SVM classifier insists on a large margin around the decision boundary. Compared to a decision hyperplane, if you have to place a fat separator between classes, you have fewer choices of where it can be put. As a result of this, the memory capacity of the model has been decreased, and hence we expect that its ability to correctly generalize to test data is increased (cf. the discussion of the bias-variance tradeoff in Chapter 14, page 312).

Let us formalize an SVM with algebra. A decision hyperplane (page 302) can be defined by an intercept term b and a decision hyperplane normal vector \vec{w} which is perpendicular to the hyperplane. This vector is commonly

WEIGHT VECTOR

referred to in the machine learning literature as the *weight vector*. To choose among all the hyperplanes that are perpendicular to the normal vector, we specify the intercept term b . Because the hyperplane is perpendicular to the normal vector, all points \vec{x} on the hyperplane satisfy $\vec{w}^\top \vec{x} = -b$. Now suppose that we have a set of training data points $\mathbb{D} = \{(\vec{x}_i, y_i)\}$, where each member is a pair of a point \vec{x}_i and a class label y_i corresponding to it.¹ For SVMs, the two data classes are always named +1 and -1 (rather than 1 and 0), and the intercept term is always explicitly represented as b (rather than being folded into the weight vector \vec{w} by adding an extra always-on feature). The math works out much more cleanly if you do things this way, as we will see almost immediately in the definition of functional margin. The linear classifier is then:

$$(15.1) \quad f(\vec{x}) = \text{sign}(\vec{w}^\top \vec{x} + b)$$

A value of -1 indicates one class, and a value of +1 the other class.

FUNCTIONAL MARGIN

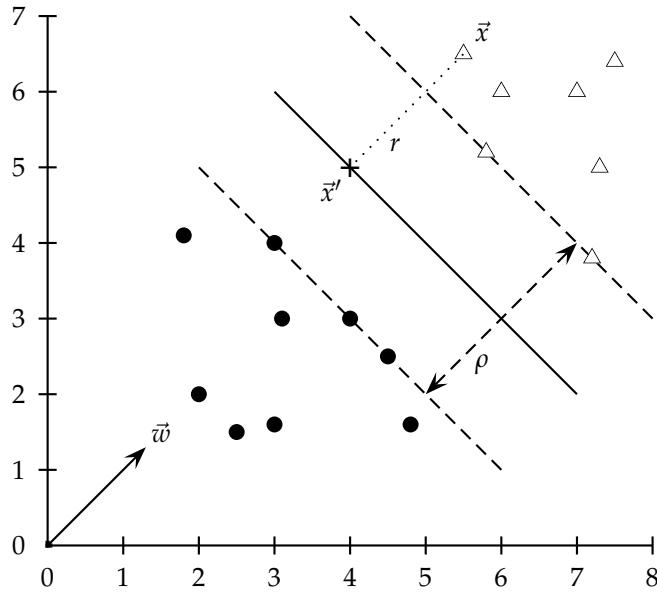
We are confident in the classification of a point if it is far away from the decision boundary. For a given data set and decision hyperplane, we define the *functional margin* of the i^{th} example \vec{x}_i with respect to a hyperplane (\vec{w}, b) as the quantity $y_i(\vec{w}^\top \vec{x}_i + b)$. The functional margin of a data set with respect to a decision surface is then twice the functional margin of any of the points in the data set with minimal functional margin (the factor of 2 comes from measuring across the whole width of the margin, as in Figure 15.3). However, there is a problem with using this definition as is: the value is underconstrained, because we can always make the functional margin as big as we wish by simply scaling up \vec{w} and b . For example, if we replace \vec{w} by $5\vec{w}$ and b by $5b$ then the functional margin $y_i(5\vec{w}^\top \vec{x}_i + 5b)$ is five times as large. This suggests that we need to place some constraint on the size of the \vec{w} vector. To get a sense of how to do that, let us look at the actual geometry.

What is the Euclidean distance from a point \vec{x} to the decision boundary? In Figure 15.3, we denote by r this distance. We know that the shortest distance between a point and a hyperplane is perpendicular to the plane, and hence, parallel to \vec{w} . A unit vector in this direction is $\vec{w}/|\vec{w}|$. The dotted line in the diagram is then a translation of the vector $r\vec{w}/|\vec{w}|$. Let us label the point on the hyperplane closest to \vec{x} as \vec{x}' . Then:

$$(15.2) \quad \vec{x}' = \vec{x} - yr \frac{\vec{w}}{|\vec{w}|}$$

where multiplying by y just changes the sign for the two cases of \vec{x} being on either side of the decision surface. Moreover, \vec{x}' lies on the decision boundary

1. As discussed in Section 14.1 (page 291), we present the general case of points in a vector space, but if the points are length normalized document vectors, then all the action is taking place on the surface of a unit sphere, and the decision surface intersects the sphere's surface.



► **Figure 15.3** The geometric margin of a point (r) and a decision boundary (ρ).

and so satisfies $\vec{w}^T \vec{x}' + b = 0$. Hence:

$$(15.3) \quad \vec{w}^T (\vec{x} - yr \frac{\vec{w}}{|\vec{w}|}) + b = 0$$

Solving for r gives:²

$$(15.4) \quad r = y \frac{\vec{w}^T \vec{x} + b}{|\vec{w}|}$$

GEOMETRIC MARGIN

Again, the points closest to the separating hyperplane are support vectors. The *geometric margin* of the classifier is the maximum width of the band that can be drawn separating the support vectors of the two classes. That is, it is twice the minimum value over data points for r given in Equation (15.4), or, equivalently, the maximal width of one of the fat separators shown in Figure 15.2. The geometric margin is clearly invariant to scaling of parameters: if we replace \vec{w} by $5\vec{w}$ and b by $5b$, then the geometric margin is the same, because it is inherently normalized by the length of \vec{w} . This means that we can impose any scaling constraint we wish on \vec{w} without affecting the geometric margin. Among other choices, we could use unit vectors, as in Chapter 6, by

2. Recall that $|\vec{w}| = \sqrt{\vec{w}^T \vec{w}}$.

requiring that $|\vec{w}| = 1$. This would have the effect of making the geometric margin the same as the functional margin.

Since we can scale the functional margin as we please, for convenience in solving large SVMs, let us choose to require that the functional margin of all data points is at least 1 and that it is equal to 1 for at least one data vector. That is, for all items in the data:

$$(15.5) \quad y_i(\vec{w}^T \vec{x}_i + b) \geq 1$$

and there exist support vectors for which the inequality is an equality. Since each example's distance from the hyperplane is $r_i = y_i(\vec{w}^T \vec{x}_i + b)/|\vec{w}|$, the geometric margin is $\rho = 2/|\vec{w}|$. Our desire is still to maximize this geometric margin. That is, we want to find \vec{w} and b such that:

- $\rho = 2/|\vec{w}|$ is maximized
- For all $(\vec{x}_i, y_i) \in \mathbb{D}$, $y_i(\vec{w}^T \vec{x}_i + b) \geq 1$

Maximizing $2/|\vec{w}|$ is the same as minimizing $|\vec{w}|/2$. This gives the final standard formulation of an SVM as a minimization problem:

$$(15.6) \quad \text{Find } \vec{w} \text{ and } b \text{ such that:}$$

- $\frac{1}{2}\vec{w}^T \vec{w}$ is minimized, and
- for all $\{(\vec{x}_i, y_i)\}$, $y_i(\vec{w}^T \vec{x}_i + b) \geq 1$

QUADRATIC PROGRAMMING

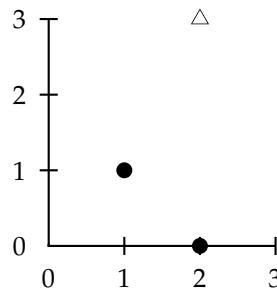
We are now optimizing a quadratic function subject to linear constraints. *Quadratic optimization* problems are a standard, well-known class of mathematical optimization problems, and many algorithms exist for solving them. We could in principle build our SVM using standard quadratic programming (QP) libraries, but there has been much recent research in this area aiming to exploit the structure of the kind of QP that emerges from an SVM. As a result, there are more intricate but much faster and more scalable libraries available especially for building SVMs, which almost everyone uses to build models. We will not present the details of such algorithms here.

However, it will be helpful to what follows to understand the shape of the solution of such an optimization problem. The solution involves constructing a dual problem where a Lagrange multiplier α_i is associated with each constraint $y_i(\vec{w}^T \vec{x}_i + b) \geq 1$ in the primal problem:

$$(15.7) \quad \text{Find } \alpha_1, \dots, \alpha_N \text{ such that } \sum \alpha_i - \frac{1}{2} \sum_i \sum_j \alpha_i \alpha_j y_i y_j \vec{x}_i^T \vec{x}_j \text{ is maximized, and}$$

- $\sum_i \alpha_i y_i = 0$
- $\alpha_i \geq 0$ for all $1 \leq i \leq N$

The solution is then of the form:



► **Figure 15.4** A tiny 3 data point training set for an SVM.

$$(15.8) \quad \vec{w} = \sum \alpha_i y_i \vec{x}_i \\ b = y_k - \vec{w}^T \vec{x}_k \text{ for any } \vec{x}_k \text{ such that } \alpha_k \neq 0$$

In the solution, most of the α_i are zero. Each non-zero α_i indicates that the corresponding \vec{x}_i is a support vector. The classification function is then:

$$(15.9) \quad f(\vec{x}) = \text{sign}(\sum_i \alpha_i y_i \vec{x}_i^T \vec{x} + b)$$

Both the term to be maximized in the dual problem and the classifying function involve a *dot product* between pairs of points (\vec{x} and \vec{x}_i or \vec{x}_i and \vec{x}_j), and that is the only way the data are used – we will return to the significance of this later.

To recap, we start with a training data set. The data set uniquely defines the best separating hyperplane, and we feed the data through a quadratic optimization procedure to find this plane. Given a new point \vec{x} to classify, the classification function $f(\vec{x})$ in either Equation (15.1) or Equation (15.9) is computing the projection of the point onto the hyperplane normal. The sign of this function determines the class to assign to the point. If the point is within the margin of the classifier (or another confidence threshold t that we might have determined to minimize classification mistakes) then the classifier can return “don’t know” rather than one of the two classes. The value of $f(\vec{x})$ may also be transformed into a probability of classification; fitting a sigmoid to transform the values is standard (Platt 2000). Also, since the margin is constant, if the model includes dimensions from various sources, careful rescaling of some dimensions may be required. However, this is not a problem if our documents (points) are on the unit hypersphere.



Example 15.1: Consider building an SVM over the (very little) data set shown in Figure 15.4. Working geometrically, for an example like this, the maximum margin weight vector will be parallel to the shortest line connecting points of the two classes, that is, the line between $(1, 1)$ and $(2, 3)$, giving a weight vector of $(1, 2)$. The optimal decision surface is orthogonal to that line and intersects it at the halfway point.

Therefore, it passes through (1.5, 2). So, the SVM decision boundary is:

$$y = x_1 + 2x_2 - 5.5$$

Working algebraically, with the standard constraint that $\text{sign}(y_i(\vec{w}^T \vec{x}_i + b)) \geq 1$, we seek to minimize $|\vec{w}|$. This happens when this constraint is satisfied with equality by the two support vectors. Further we know that the solution is $\vec{w} = (a, 2a)$ for some a . So we have that:

$$\begin{aligned} a + 2a + b &= -1 \\ 2a + 6a + b &= 1 \end{aligned}$$

Therefore, $a = 2/5$ and $b = -11/5$. So the optimal hyperplane is given by $\vec{w} = (2/5, 4/5)$ and $b = -11/5$.

The margin ρ is $2/|\vec{w}| = 2/\sqrt{4/25 + 16/25} = 2/(2\sqrt{5}/5) = \sqrt{5}$. This answer can be confirmed geometrically by examining Figure 15.4.



Exercise 15.1

[★]

What is the minimum number of support vectors that there can be for a data set (which contains instances of each class)?

Exercise 15.2

[★★]

The basis of being able to use kernels in SVMs (see Section 15.2.3) is that the classification function can be written in the form of Equation (15.9) (where, for large problems, most α_i are 0). Show explicitly how the classification function could be written in this form for the data set from Example 15.1. That is, write f as a function where the data points appear and the only variable is \vec{x} .

Exercise 15.3

[★★]

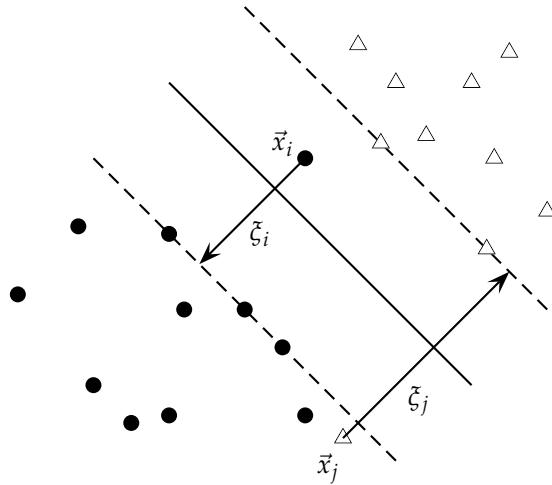
Install an SVM package such as SVMlight (<http://svmlight.joachims.org/>), and build an SVM for the data set discussed in Example 15.1. Confirm that the program gives the same solution as the text. For SVMlight, or another package that accepts the same training data format, the training file would be:

```
+1 1:2 2:3
-1 1:2 2:0
-1 1:1 2:1
```

The training command for SVMlight is then:

```
svm_learn -c 1 -a alphas.dat train.dat model.dat
```

The `-c 1` option is needed to turn off use of the slack variables that we discuss in Section 15.2.1. Check that the norm of the weight vector agrees with what we found in Example 15.1. Examine the file `alphas.dat` which contains the α_i values, and check that they agree with your answers in Exercise 15.2.



► **Figure 15.5** Large margin classification with slack variables.

15.2 Extensions to the SVM model

15.2.1 Soft margin classification

For the very high dimensional problems common in text classification, sometimes the data are linearly separable. But in the general case they are not, and even if they are, we might prefer a solution that better separates the bulk of the data while ignoring a few weird noise documents.

If the training set \mathbb{D} is not linearly separable, the standard approach is to allow the fat decision margin to make a few mistakes (some points – outliers or noisy examples – are inside or on the wrong side of the margin). We then pay a cost for each misclassified example, which depends on how far it is from meeting the margin requirement given in Equation (15.5). To implement this, we introduce *slack variables* ξ_i . A non-zero value for ξ_i allows \vec{x}_i to not meet the margin requirement at a cost proportional to the value of ξ_i . See Figure 15.5.

The formulation of the SVM optimization problem with slack variables is:

(15.10) Find \vec{w} , b , and $\xi_i \geq 0$ such that:

- $\frac{1}{2}\vec{w}^T\vec{w} + C \sum_i \xi_i$ is minimized
- and for all $\{(\vec{x}_i, y_i)\}$, $y_i(\vec{w}^T\vec{x}_i + b) \geq 1 - \xi_i$

REGULARIZATION

The optimization problem is then trading off how fat it can make the margin versus how many points have to be moved around to allow this margin. The margin can be less than 1 for a point \vec{x}_i by setting $\xi_i > 0$, but then one pays a penalty of $C\xi_i$ in the minimization for having done that. The sum of the ξ_i gives an upper bound on the number of training errors. Soft-margin SVMs minimize training error traded off against margin. The parameter C is a *regularization* term, which provides a way to control overfitting: as C becomes large, it is unattractive to not respect the data at the cost of reducing the geometric margin; when it is small, it is easy to account for some data points with the use of slack variables and to have a fat margin placed so it models the bulk of the data.

The dual problem for soft margin classification becomes:

- (15.11) Find $\alpha_1, \dots, \alpha_N$ such that $\sum \alpha_i - \frac{1}{2} \sum_i \sum_j \alpha_i \alpha_j y_i y_j \vec{x}_i^T \vec{x}_j$ is maximized, and
- $\sum_i \alpha_i y_i = 0$
 - $0 \leq \alpha_i \leq C$ for all $1 \leq i \leq N$

Neither the slack variables ξ_i nor Lagrange multipliers for them appear in the dual problem. All we are left with is the constant C bounding the possible size of the Lagrange multipliers for the support vector data points. As before, the \vec{x}_i with non-zero α_i will be the support vectors. The solution of the dual problem is of the form:

$$(15.12) \quad \begin{aligned} \vec{w} &= \sum \alpha_i y_i \vec{x}_i \\ b &= y_k(1 - \xi_k) - \vec{w}^T \vec{x}_k \text{ for } k = \arg \max_k \alpha_k \end{aligned}$$

Again \vec{w} is not needed explicitly for classification, which can be done in terms of dot products with data points, as in Equation (15.9).

Typically, the support vectors will be a small proportion of the training data. However, if the problem is non-separable or with small margin, then every data point which is misclassified or within the margin will have a non-zero α_i . If this set of points becomes large, then, for the nonlinear case which we turn to in Section 15.2.3, this can be a major slowdown for using SVMs at test time.

The complexity of training and testing with linear SVMs is shown in Table 15.1.³ The time for training an SVM is dominated by the time for solving the underlying QP, and so the theoretical and empirical complexity varies depending on the method used to solve it. The standard result for solving QPs is that it takes time cubic in the size of the data set (Kozlov et al. 1979). All the recent work on SVM training has worked to reduce that complexity, often by

3. We write $\Theta(|\mathcal{D}|_{\text{ave}})$ for $\Theta(T)$ (page 262) and assume that the length of test documents is bounded as we did on page 262.

Classifier	Mode	Method	Time complexity
NB	training		$\Theta(\mathcal{D} L_{ave} + \mathcal{C} V)$
NB	testing		$\Theta(\mathcal{C} M_a)$
Rocchio	training		$\Theta(\mathcal{D} L_{ave} + \mathcal{C} V)$
Rocchio	testing		$\Theta(\mathcal{C} M_a)$
kNN	training	preprocessing	$\Theta(\mathcal{D} L_{ave})$
kNN	testing	preprocessing	$\Theta(\mathcal{D} M_{ave}M_a)$
kNN	training	no preprocessing	$\Theta(1)$
kNN	testing	no preprocessing	$\Theta(\mathcal{D} L_{ave}M_a)$
SVM	training	conventional	$O(\mathcal{C} \mathcal{D} ^3 M_{ave});$ $\approx O(\mathcal{C} \mathcal{D} ^{1.7} M_{ave})$, empirically
SVM	training	cutting planes	$O(\mathcal{C} \mathcal{D} M_{ave})$
SVM	testing		$O(\mathcal{C} M_a)$

► **Table 15.1** Training and testing complexity of various classifiers including SVMs. Training is the time the learning method takes to learn a classifier over \mathcal{D} , while testing is the time it takes a classifier to classify one document. For SVMs, multiclass classification is assumed to be done by a set of $|\mathcal{C}|$ one-versus-rest classifiers. L_{ave} is the average number of tokens per document, while M_{ave} is the average vocabulary (number of non-zero features) of a document. L_a and M_a are the numbers of tokens and types, respectively, in the test document.

being satisfied with approximate solutions. Standardly, empirical complexity is about $O(|\mathcal{D}|^{1.7})$ (Joachims 2006a). Nevertheless, the super-linear training time of traditional SVM algorithms makes them difficult or impossible to use on very large training data sets. Alternative traditional SVM solution algorithms which are linear in the number of training examples scale badly with a large number of features, which is another standard attribute of text problems. However, a new training algorithm based on cutting plane techniques gives a promising answer to this issue by having running time linear in the number of training examples and the number of non-zero features in examples (Joachims 2006a). Nevertheless, the actual speed of doing quadratic optimization remains much slower than simply counting terms as is done in a Naive Bayes model. Extending SVM algorithms to nonlinear SVMs, as in the next section, standardly increases training complexity by a factor of $|\mathcal{D}|$ (since dot products between examples need to be calculated), making them impractical. In practice it can often be cheaper to materialize

the higher-order features and to train a linear SVM.⁴

15.2.2 Multiclass SVMs

SVMs are inherently two-class classifiers. The traditional way to do multiclass classification with SVMs is to use one of the methods discussed in Section 14.5 (page 306). In particular, the most common technique in practice has been to build $|C|$ one-versus-rest classifiers (commonly referred to as “one-versus-all” or OVA classification), and to choose the class which classifies the test datum with greatest margin. Another strategy is to build a set of one-versus-one classifiers, and to choose the class that is selected by the most classifiers. While this involves building $|C|(|C| - 1)/2$ classifiers, the time for training classifiers may actually decrease, since the training data set for each classifier is much smaller.

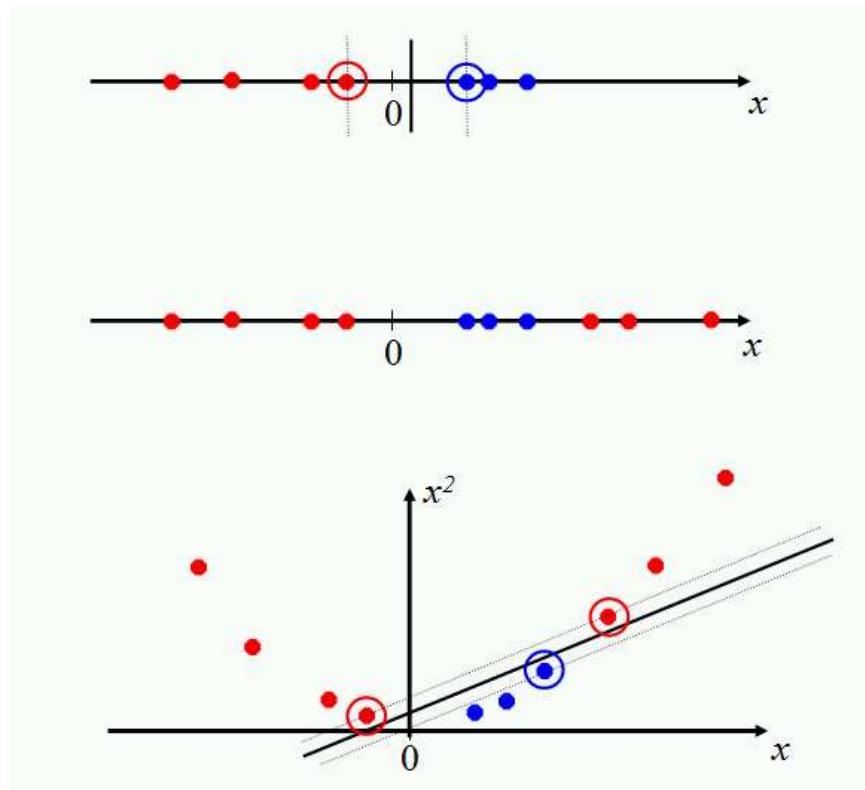
However, these are not very elegant approaches to solving multiclass problems. A better alternative is provided by the construction of multiclass SVMs, where we build a two-class classifier over a feature vector $\Phi(\vec{x}, y)$ derived from the pair consisting of the input features and the class of the datum. At test time, the classifier chooses the class $y = \arg \max_{y'} \vec{w}^T \Phi(\vec{x}, y')$. The margin during training is the gap between this value for the correct class and for the nearest other class, and so the quadratic program formulation will require that $\forall i \forall y \neq y_i \vec{w}^T \Phi(\vec{x}_i, y_i) - \vec{w}^T \Phi(\vec{x}_i, y) \geq 1 - \xi_i$. This general method can be extended to give a multiclass formulation of various kinds of linear classifiers. It is also a simple instance of a generalization of classification where the classes are not just a set of independent, categorical labels, but may be arbitrary structured objects with relationships defined between them. In the SVM world, such work comes under the label of *structural SVMs*. We mention them again in Section 15.4.2.

STRUCTURAL SVMs

15.2.3 Nonlinear SVMs

With what we have presented so far, data sets that are linearly separable (perhaps with a few exceptions or some noise) are well-handled. But what are we going to do if the data set just doesn’t allow classification by a linear classifier? Let us look at a one-dimensional case. The top data set in Figure 15.6 is straightforwardly classified by a linear classifier but the middle data set is not. We instead need to be able to pick out an interval. One way to solve this problem is to map the data on to a higher dimensional space and then to use a linear classifier in the higher dimensional space. For example, the bottom part of the figure shows that a linear separator can easily classify the data

⁴. Materializing the features refers to directly calculating higher order and interaction terms and then putting them into a linear model.



► **Figure 15.6** Projecting data that is not linearly separable into a higher dimensional space can make it linearly separable.

if we use a quadratic function to map the data into two dimensions (a polar coordinates projection would be another possibility). The general idea is to map the original feature space to some higher-dimensional feature space where the training set is separable. Of course, we would want to do so in ways that preserve relevant dimensions of relatedness between data points, so that the resultant classifier should still generalize well.

KERNEL TRICK

SVMs, and also a number of other linear classifiers, provide an easy and efficient way of doing this mapping to a higher dimensional space, which is referred to as “the *kernel trick*”. It’s not really a trick: it just exploits the math that we have seen. The SVM linear classifier relies on a dot product between data point vectors. Let $K(\vec{x}_i, \vec{x}_j) = \vec{x}_i^T \vec{x}_j$. Then the classifier we have seen so

KERNEL FUNCTION

far is:

$$(15.13) \quad f(\vec{x}) = \text{sign}(\sum_i \alpha_i y_i K(\vec{x}_i, \vec{x}) + b)$$

Now suppose we decide to map every data point into a higher dimensional space via some transformation $\Phi: \vec{x} \mapsto \phi(\vec{x})$. Then the dot product becomes $\phi(\vec{x}_i)^T \phi(\vec{x}_j)$. If it turned out that this dot product (which is just a real number) could be computed simply and efficiently in terms of the original data points, then we wouldn't have to actually map from $\vec{x} \mapsto \phi(\vec{x})$. Rather, we could simply compute the quantity $K(\vec{x}_i, \vec{x}_j) = \phi(\vec{x}_i)^T \phi(\vec{x}_j)$, and then use the function's value in Equation (15.13). A *kernel function* K is such a function that corresponds to a dot product in some expanded feature space.



Example 15.2: The quadratic kernel in two dimensions. For 2-dimensional vectors $\vec{u} = (u_1 \ u_2)$, $\vec{v} = (v_1 \ v_2)$, consider $K(\vec{u}, \vec{v}) = (1 + \vec{u}^T \vec{v})^2$. We wish to show that this is a kernel, i.e., that $K(\vec{u}, \vec{v}) = \phi(\vec{u})^T \phi(\vec{v})$ for some ϕ . Consider $\phi(\vec{u}) = (1 \ u_1^2 \ \sqrt{2}u_1u_2 \ u_2^2 \ \sqrt{2}u_1 \ \sqrt{2}u_2)$. Then:

$$(15.14) \quad \begin{aligned} K(\vec{u}, \vec{v}) &= (1 + \vec{u}^T \vec{v})^2 \\ &= 1 + u_1^2 v_1^2 + 2u_1 v_1 u_2 v_2 + u_2^2 v_2^2 + 2u_1 v_1 + 2u_2 v_2 \\ &= (1 \ u_1^2 \ \sqrt{2}u_1u_2 \ u_2^2 \ \sqrt{2}u_1 \ \sqrt{2}u_2)^T (1 \ v_1^2 \ \sqrt{2}v_1v_2 \ v_2^2 \ \sqrt{2}v_1 \ \sqrt{2}v_2) \\ &= \phi(\vec{u})^T \phi(\vec{v}) \end{aligned}$$

KERNEL
MERCER KERNEL

In the language of functional analysis, what kinds of functions are valid *kernel functions*? Kernel functions are sometimes more precisely referred to as *Mercer kernels*, because they must satisfy Mercer's condition: for any $g(\vec{x})$ such that $\int g(\vec{x})^2 d\vec{x}$ is finite, we must have that:

$$(15.15) \quad \int K(\vec{x}, \vec{z}) g(\vec{x}) g(\vec{z}) d\vec{x} d\vec{z} \geq 0.$$

A kernel function K must be continuous, symmetric, and have a positive definite gram matrix. Such a K means that there exists a mapping to a reproducing kernel Hilbert space (a Hilbert space is a vector space closed under dot products) such that the dot product there gives the same value as the function K . If a kernel does not satisfy Mercer's condition, then the corresponding QP may have no solution. If you would like to better understand these issues, you should consult the books on SVMs mentioned in Section 15.5. Otherwise, you can content yourself with knowing that 90% of work with kernels uses one of two straightforward families of functions of two vectors, which we define below, and which define valid kernels.

The two commonly used families of kernels are polynomial kernels and radial basis functions. Polynomial kernels are of the form $K(\vec{x}, \vec{z}) = (1 +$

$\vec{x}^T \vec{z})^d$. The case of $d = 1$ is a linear kernel, which is what we had before the start of this section (the constant 1 just changing the threshold). The case of $d = 2$ gives a quadratic kernel, and is very commonly used. We illustrated the quadratic kernel in Example 15.2.

The most common form of radial basis function is a Gaussian distribution, calculated as:

$$(15.16) \quad K(\vec{x}, \vec{z}) = e^{-(\vec{x}-\vec{z})^2/(2\sigma^2)}$$

A radial basis function (rbf) is equivalent to mapping the data into an infinite dimensional Hilbert space, and so we cannot illustrate the radial basis function concretely, as we did a quadratic kernel. Beyond these two families, there has been interesting work developing other kernels, some of which is promising for text applications. In particular, there has been investigation of string kernels (see Section 15.5).

The world of SVMs comes with its own language, which is rather different from the language otherwise used in machine learning. The terminology does have deep roots in mathematics, but it's important not to be too awed by that terminology. Really, we are talking about some quite simple things. A polynomial kernel allows us to model feature conjunctions (up to the order of the polynomial). That is, if we want to be able to model occurrences of pairs of words, which give distinctive information about topic classification, not given by the individual words alone, like perhaps *operating AND system* or *ethnic AND cleansing*, then we need to use a quadratic kernel. If occurrences of triples of words give distinctive information, then we need to use a cubic kernel. Simultaneously you also get the powers of the basic features – for most text applications, that probably isn't useful, but just comes along with the math and hopefully doesn't do harm. A radial basis function allows you to have features that pick out circles (hyperspheres) – although the decision boundaries become much more complex as multiple such features interact. A string kernel lets you have features that are character subsequences of terms. All of these are straightforward notions which have also been used in many other places under different names.

15.2.4 Experimental results

We presented results in Section 13.6 showing that an SVM is a very effective text classifier. The results of Dumais et al. (1998) given in Table 13.9 show SVMs clearly performing the best. This was one of several pieces of work from this time that established the strong reputation of SVMs for text classification. Another pioneering work on scaling and evaluating SVMs for text classification was (Joachims 1998). We present some of his results

	NB	Roc-	Dec.	kNN	linear SVM	rbf-SVM
		chio	Trees		C = 0.5	C = 1.0
earn	96.0	96.1	96.1	97.8	98.0	98.2
acq	90.7	92.1	85.3	91.8	95.5	95.6
money-fx	59.6	67.6	69.4	75.4	78.8	78.5
grain	69.8	79.5	89.1	82.6	91.9	93.1
crude	81.2	81.5	75.5	85.8	89.4	88.7
trade	52.2	77.4	59.2	77.9	79.2	79.2
interest	57.6	72.5	49.1	76.7	75.6	74.8
ship	80.9	83.1	80.9	79.8	87.4	86.5
wheat	63.4	79.4	85.5	72.9	86.6	86.8
corn	45.2	62.2	87.7	71.4	87.5	87.8
microavg.	72.3	79.9	79.4	82.6	86.7	87.5
						86.4

► **Table 15.2** SVM classifier break-even F_1 from (Joachims 2002a, p. 114). Results are shown for the 10 largest categories and for microaveraged performance over all 90 categories on the Reuters-21578 data set.

from (Joachims 2002a) in Table 15.2.⁵ Joachims used a large number of term features in contrast to Dumais et al. (1998), who used MI feature selection (Section 13.5.1, page 272) to build classifiers with a much more limited number of features. The success of the linear SVM mirrors the results discussed in Section 14.6 (page 308) on other linear approaches like Naive Bayes. It seems that working with simple term features can get one a long way. It is again noticeable the extent to which different papers' results for the same machine learning methods differ. In particular, based on replications by other researchers, the Naive Bayes results of (Joachims 1998) appear too weak, and the results in Table 13.9 should be taken as representative.

15.3 Issues in the classification of text documents

There are lots of applications of text classification in the commercial world; email spam filtering is perhaps now the most ubiquitous. Jackson and Moulinier (2002) write: “There is no question concerning the commercial value of being able to classify documents automatically by content. There are myriad

5. These results are in terms of the break-even F_1 (see Section 8.4). Many researchers disprefer this measure for text classification evaluation, since its calculation may involve interpolation rather than an actual parameter setting of the system and it is not clear why this value should be reported rather than maximal F_1 or another point on the precision/recall curve motivated by the task at hand. While earlier results in (Joachims 1998) suggested notable gains on this task from the use of higher order polynomial or rbf kernels, this was with hard-margin SVMs. With soft-margin SVMs, a simple linear SVM with the default $C = 1$ performs best.

potential applications of such a capability for corporate Intranets, government departments, and Internet publishers.”

Most of our discussion of classification has focused on introducing various machine learning methods rather than discussing particular features of text documents relevant to classification. This bias is appropriate for a textbook, but is misplaced for an application developer. It is frequently the case that greater performance gains can be achieved from exploiting domain-specific text features than from changing from one machine learning method to another. Jackson and Moulinier (2002) suggest that “Understanding the data is one of the keys to successful categorization, yet this is an area in which most categorization tool vendors are extremely weak. Many of the ‘one size fits all’ tools on the market have not been tested on a wide range of content types.” In this section we wish to step back a little and consider the applications of text classification, the space of possible solutions, and the utility of application-specific heuristics.

15.3.1 Choosing what kind of classifier to use

When confronted with a need to build a text classifier, the first question to ask is how much training data is there currently available? None? Very little? Quite a lot? Or a huge amount, growing every day? Often one of the biggest practical challenges in fielding a machine learning classifier in real applications is creating or obtaining enough training data. For many problems and algorithms, hundreds or thousands of examples from each class are required to produce a high performance classifier and many real world contexts involve large sets of categories. We will initially assume that the classifier is needed as soon as possible; if a lot of time is available for implementation, much of it might be spent on assembling data resources.

If you have no labeled training data, and especially if there are existing staff knowledgeable about the domain of the data, then you should never forget the solution of using hand-written rules. That is, you write standing queries, as we touched on at the beginning of Chapter 13. For example:

IF (wheat OR grain) AND NOT (whole OR bread) THEN $c = \text{grain}$

In practice, rules get a lot bigger than this, and can be phrased using more sophisticated query languages than just Boolean expressions, including the use of numeric scores. With careful crafting (that is, by humans tuning the rules on development data), the accuracy of such rules can become very high. Jacobs and Rau (1990) report identifying articles about takeovers with 92% precision and 88.5% recall, and Hayes and Weinstein (1990) report 94% recall and 84% precision over 675 categories on Reuters newswire documents. Nevertheless the amount of work to create such well-tuned rules is very large. A reasonable estimate is 2 days per class, and extra time has to go

into maintenance of rules, as the content of documents in classes drifts over time (cf. page 269).

If you have fairly little data and you are going to train a supervised classifier, then machine learning theory says you should stick to a classifier with high bias, as we discussed in Section 14.6 (page 308). For example, there are theoretical and empirical results that Naive Bayes does well in such circumstances (Ng and Jordan 2001, Forman and Cohen 2004), although this effect is not necessarily observed in practice with regularized models over textual data (Klein and Manning 2002). At any rate, a very low bias model like a nearest neighbor model is probably counterindicated. Regardless, the quality of the model will be adversely affected by the limited training data.

Here, the theoretically interesting answer is to try to apply *semi-supervised training methods*. This includes methods such as bootstrapping or the EM algorithm, which we will introduce in Section 16.5 (page 368). In these methods, the system gets some labeled documents, and a further large supply of unlabeled documents over which it can attempt to learn. One of the big advantages of Naive Bayes is that it can be straightforwardly extended to be a semi-supervised learning algorithm, but for SVMs, there is also semi-supervised learning work which goes under the title of *transductive SVMs*. See the references for pointers.

Often, the practical answer is to work out how to get more labeled data as quickly as you can. The best way to do this is to insert yourself into a process where humans will be willing to label data for you as part of their natural tasks. For example, in many cases humans will sort or route email for their own purposes, and these actions give information about classes. The alternative of getting human labelers expressly for the task of training classifiers is often difficult to organize, and the labeling is often of lower quality, because the labels are not embedded in a realistic task context. Rather than getting people to label all or a random sample of documents, there has also been considerable research on *active learning*, where a system is built which decides which documents a human should label. Usually these are the ones on which a classifier is uncertain of the correct classification. This can be effective in reducing annotation costs by a factor of 2–4, but has the problem that the good documents to label to train one type of classifier often are not the good documents to label to train a different type of classifier.

If there is a reasonable amount of labeled data, then you are in the perfect position to use everything that we have presented about text classification. For instance, you may wish to use an SVM. However, if you are deploying a linear classifier such as an SVM, you should probably design an application that overlays a Boolean rule-based classifier over the machine learning classifier. Users frequently like to adjust things that do not come out quite right, and if management gets on the phone and wants the classification of a particular document fixed right now, then this is much easier to

SEMI-SUPERVISED
LEARNING

TRANSDUCTIVE SVMs

ACTIVE LEARNING

do by hand-writing a rule than by working out how to adjust the weights of an SVM without destroying the overall classification accuracy. This is one reason why machine learning models like decision trees which produce user-interpretable Boolean-like models retain considerable popularity.

If a huge amount of data are available, then the choice of classifier probably has little effect on your results and the best choice may be unclear (cf. [Banko and Brill 2001](#)). It may be best to choose a classifier based on the scalability of training or even runtime efficiency. To get to this point, you need to have huge amounts of data. The general rule of thumb is that each doubling of the training data size produces a linear increase in classifier performance, but with very large amounts of data, the improvement becomes sub-linear.

15.3.2 Improving classifier performance

For any particular application, there is usually significant room for improving classifier effectiveness through exploiting features specific to the domain or document collection. Often documents will contain zones which are especially useful for classification. Often there will be particular subvocabularies which demand special treatment for optimal classification effectiveness.

Large and difficult category taxonomies

If a text classification problem consists of a small number of well-separated categories, then many classification algorithms are likely to work well. But many real classification problems consist of a very large number of often very similar categories. The reader might think of examples like web directories (the Yahoo! Directory or the Open Directory Project), library classification schemes (Dewey Decimal or Library of Congress) or the classification schemes used in legal or medical applications. For instance, the Yahoo! Directory consists of over 200,000 categories in a deep hierarchy. Accurate classification over large sets of closely related classes is inherently difficult.

Most large sets of categories have a hierarchical structure, and attempting to exploit the hierarchy by doing *hierarchical classification* is a promising approach. However, at present the effectiveness gains from doing this rather than just working with the classes that are the leaves of the hierarchy remain modest.⁶ But the technique can be very useful simply to improve the scalability of building classifiers over large hierarchies. Another simple way to improve the scalability of classifiers over large hierarchies is the use of aggressive feature selection. We provide references to some work on hierarchical classification in Section 15.5.

HIERARCHICAL
CLASSIFICATION

6. Using the small hierarchy in Figure 13.1 (page 257) as an example, the leaf classes are ones like *poultry* and *coffee*, as opposed to higher-up classes like *industries*.

A general result in machine learning is that you can always get a small boost in classification accuracy by combining multiple classifiers, provided only that the mistakes that they make are at least somewhat independent. There is now a large literature on techniques such as voting, bagging, and boosting multiple classifiers. Again, there are some pointers in the references. Nevertheless, ultimately a hybrid automatic/manual solution may be needed to achieve sufficient classification accuracy. A common approach in such situations is to run a classifier first, and to accept all its high confidence decisions, but to put low confidence decisions in a queue for manual review. Such a process also automatically leads to the production of new training data which can be used in future versions of the machine learning classifier. However, note that this is a case in point where the resulting training data is clearly *not* randomly sampled from the space of documents.

Features for text

The default in both ad hoc retrieval and text classification is to use terms as features. However, for text classification, a great deal of mileage can be achieved by designing additional features which are suited to a specific problem. Unlike the case of IR query languages, since these features are internal to the classifier, there is no problem of communicating these features to an end user. This process is generally referred to as *feature engineering*. At present, feature engineering remains a human craft, rather than something done by machine learning. Good feature engineering can often markedly improve the performance of a text classifier. It is especially beneficial in some of the most important applications of text classification, like spam and porn filtering.

Classification problems will often contain large numbers of terms which can be conveniently grouped, and which have a similar vote in text classification problems. Typical examples might be year mentions or strings of exclamation marks. Or they may be more specialized tokens like ISBNs or chemical formulas. Often, using them directly in a classifier would greatly increase the vocabulary without providing classificatory power beyond knowing that, say, a chemical formula is present. In such cases, the number of features and feature sparseness can be reduced by matching such items with regular expressions and converting them into distinguished tokens. Consequently, effectiveness and classifier speed are normally enhanced. Sometimes all numbers are converted into a single feature, but often some value can be had by distinguishing different kinds of numbers, such as four digit numbers (which are usually years) versus other cardinal numbers versus real numbers with a decimal point. Similar techniques can be applied to dates, ISBN numbers, sports game scores, and so on.

Going in the other direction, it is often useful to increase the number of fea-

tures by matching parts of words, and by matching selected multiword patterns that are particularly discriminative. Parts of words are often matched by character k -gram features. Such features can be particularly good at providing classification clues for otherwise unknown words when the classifier is deployed. For instance, an unknown word ending in *-rase* is likely to be an enzyme, even if it wasn't seen in the training data. Good multiword patterns are often found by looking for distinctively common word pairs (perhaps using a mutual information criterion between words, in a similar way to its use in Section 13.5.1 (page 272) for feature selection) and then using feature selection methods evaluated against classes. They are useful when the components of a compound would themselves be misleading as classification cues. For instance, this would be the case if the keyword *ethnic* was most indicative of the categories *food* and *arts*, the keyword *cleansing* was most indicative of the category *home*, but the collocation *ethnic cleansing* instead indicates the category *world news*. Some text classifiers also make use of features from named entity recognizers (cf. page 195).

Do techniques like stemming and lowercasing (Section 2.2, page 22) help for text classification? As always, the ultimate test is empirical evaluations conducted on an appropriate test collection. But it is nevertheless useful to note that such techniques have a more restricted chance of being useful for classification. For IR, you often need to collapse forms of a word like *oxy-generate* and *oxygenation*, because the appearance of either in a document is a good clue that the document will be relevant to a query about oxygenation. Given copious training data, stemming necessarily delivers no value for text classification. If several forms that stem together have a similar signal, the parameters estimated for all of them will have similar weights. Techniques like stemming help only in compensating for data sparseness. This can be a useful role (as noted at the start of this section), but often different forms of a word can convey significantly different cues about the correct document classification. Overly aggressive stemming can easily degrade classification performance.

Document zones in text classification

As already discussed in Section 6.1, documents usually have zones, such as mail message headers like the subject and author, or the title and keywords of a research article. Text classifiers can usually gain from making use of these zones during training and classification.

Upweighting document zones. In text classification problems, you can frequently get a nice boost to effectiveness by differentially weighting contributions from different document zones. Often, upweighting title words is particularly effective (Cohen and Singer 1999, p. 163). As a rule of thumb,

it is often effective to double the weight of title words in text classification problems. You can also get value from upweighting words from pieces of text that are not so much clearly defined zones, but where nevertheless evidence from document structure or content suggests that they are important. [Murata et al. \(2000\)](#) suggest that you can also get value (in an ad hoc retrieval context) from upweighting the first sentence of a (newswire) document.

Separate feature spaces for document zones. There are two strategies that can be used for document zones. Above we upweighted words that appear in certain zones. This means that we are using the same features (that is, parameters are “tied” across different zones), but we pay more attention to the occurrence of terms in particular zones. An alternative strategy is to have a completely separate set of features and corresponding parameters for words occurring in different zones. This is in principle more powerful: a word could usually indicate the topic *Middle East* when in the title but *Commodities* when in the body of a document. But, in practice, tying parameters is usually more successful. Having separate feature sets means having two or more times as many parameters, many of which will be much more sparsely seen in the training data, and hence with worse estimates, whereas upweighting has no bad effects of this sort. Moreover, it is quite uncommon for words to have different preferences when appearing in different zones; it is mainly the strength of their vote that should be adjusted. Nevertheless, ultimately this is a contingent result, depending on the nature and quantity of the training data.

Connections to text summarization. In Section 8.7, we mentioned the field of text summarization, and how most work in that field has adopted the limited goal of extracting and assembling pieces of the original text that are judged to be central based on features of sentences that consider the sentence’s position and content. Much of this work can be used to suggest zones that may be distinctively useful for text classification. For example [Kołcz et al. \(2000\)](#) consider a form of feature selection where you classify documents based only on words in certain zones. Based on text summarization research, they consider using (i) only the title, (ii) only the first paragraph, (iii) only the paragraph with the most title words or keywords, (iv) the first two paragraphs or the first and last paragraph, or (v) all sentences with a minimum number of title words or keywords. In general, these positional feature selection methods produced as good results as mutual information (Section 13.5.1), and resulted in quite competitive classifiers. [Ko et al. \(2004\)](#) also took inspiration from text summarization research to upweight sentences with either words from the title or words that are central to the document’s content, leading to classification accuracy gains of almost 1%. This

presumably works because most such sentences are somehow more central to the concerns of the document.


Exercise 15.4

[**]

Spam email often makes use of various cloaking techniques to try to get through. One method is to pad or substitute characters so as to defeat word-based text classifiers. For example, you see terms like the following in spam email:

Rep1icaRolex	bonmus	Viiiaaaaagra	pi11z
PHARlbdMACY	[LEV]i[IT]l[RA]	se\exual	ClAfLIS

Discuss how you could engineer features that would largely defeat this strategy.

Exercise 15.5

[**]

Another strategy often used by purveyors of email spam is to follow the message they wish to send (such as buying a cheap stock or whatever) with a paragraph of text from another innocuous source (such as a news article). Why might this strategy be effective? How might it be addressed by a text classifier?

Exercise 15.6

[*]

What other kinds of features appear as if they would be useful in an email spam classifier?

15.4 Machine learning methods in ad hoc information retrieval

Rather than coming up with term and document weighting functions by hand, as we primarily did in Chapter 6, we can view different sources of relevance signal (cosine score, title match, etc.) as features in a learning problem. A classifier that has been fed examples of relevant and nonrelevant documents for each of a set of queries can then figure out the relative weights of these signals. If we configure the problem so that there are pairs of a document and a query which are assigned a relevance judgment of *relevant* or *nonrelevant*, then we can think of this problem too as a text classification problem. Taking such a classification approach is not necessarily best, and we present an alternative in Section 15.4.2. Nevertheless, given the material we have covered, the simplest place to start is to approach this problem as a classification problem, by ordering the documents according to the confidence of a two-class classifier in its relevance decision. And this move is not purely pedagogical; exactly this approach is sometimes used in practice.

15.4.1 A simple example of machine-learned scoring

In this section we generalize the methodology of Section 6.1.2 (page 113) to *machine learning* of the scoring function. In Section 6.1.2 we considered a case where we had to combine Boolean indicators of relevance; here we consider more general factors to further develop the notion of machine-learned

Example	DocID	Query	Cosine score	ω	Judgment
Φ_1	37	linux operating system	0.032	3	relevant
Φ_2	37	penguin logo	0.02	4	nonrelevant
Φ_3	238	operating system	0.043	2	relevant
Φ_4	238	runtime environment	0.004	2	nonrelevant
Φ_5	1741	kernel layer	0.022	3	relevant
Φ_6	2094	device driver	0.03	2	relevant
Φ_7	3191	device driver	0.027	5	nonrelevant
...

► **Table 15.3** Training examples for machine-learned scoring.

relevance. In particular, the factors we now consider go beyond Boolean functions of query term presence in document zones, as in Section 6.1.2.

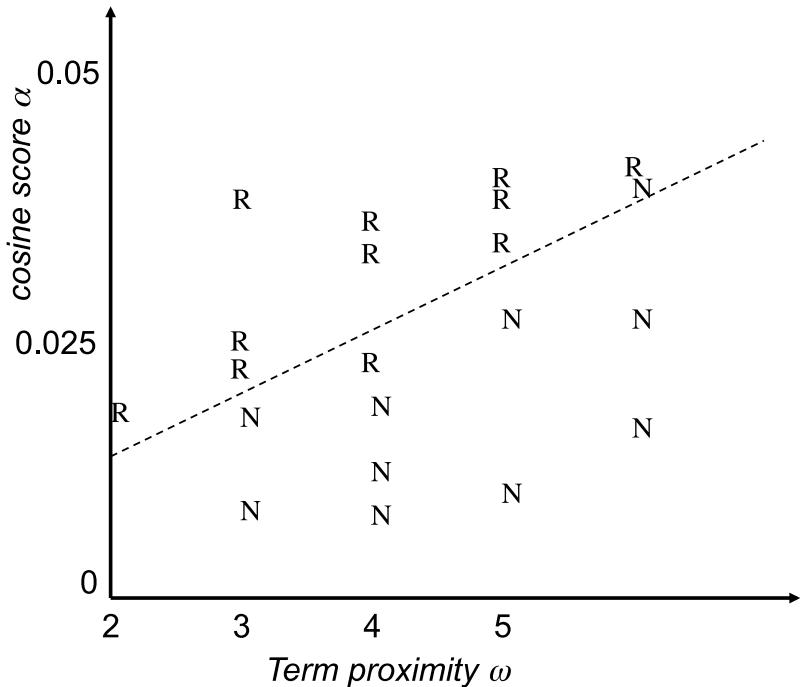
We develop the ideas in a setting where the scoring function is a linear combination of two factors: (1) the vector space cosine similarity between query and document and (2) the minimum window width ω within which the query terms lie. As we noted in Section 7.2.2 (page 144), query term proximity is often very indicative of a document being on topic, especially with longer documents and on the web. Among other things, this quantity gives us an implementation of implicit phrases. Thus we have one factor that depends on the statistics of query terms in the document as a bag of words, and another that depends on proximity weighting. We consider only two features in the development of the ideas because a two-feature exposition remains simple enough to visualize. The technique can be generalized to many more features.

As in Section 6.1.2, we are provided with a set of *training examples*, each of which is a pair consisting of a query and a document, together with a relevance judgment for that document on that query that is either *relevant* or *nonrelevant*. For each such example we can compute the vector space cosine similarity, as well as the window width ω . The result is a training set as shown in Table 15.3, which resembles Figure 6.5 (page 115) from Section 6.1.2.

Here, the two features (cosine score denoted α and window width ω) are real-valued predictors. If we once again quantify the judgment *relevant* as 1 and *nonrelevant* as 0, we seek a scoring function that combines the values of the features to generate a value that is (close to) 0 or 1. We wish this function to be in agreement with our set of training examples as far as possible. Without loss of generality, a linear classifier will use a linear combination of features of the form

$$(15.17) \quad Score(d, q) = Score(\alpha, \omega) = a\alpha + b\omega + c,$$

with the coefficients a, b, c to be learned from the training data. While it is



► **Figure 15.7** A collection of training examples. Each R denotes a training example labeled *relevant*, while each N is a training example labeled *nonrelevant*.

possible to formulate this as an error minimization problem as we did in Section 6.1.2, it is instructive to visualize the geometry of Equation (15.17). The examples in Table 15.3 can be plotted on a two-dimensional plane with axes corresponding to the cosine score α and the window width ω . This is depicted in Figure 15.7.

In this setting, the function $Score(\alpha, \omega)$ from Equation (15.17) represents a plane “hanging above” Figure 15.7. Ideally this plane (in the direction perpendicular to the page containing Figure 15.7) assumes values close to 1 above the points marked R, and values close to 0 above the points marked N. Since a plane is unlikely to assume only values close to 0 or 1 above the training sample points, we make use of *thresholding*: given any query and document for which we wish to determine relevance, we pick a value θ and if $Score(\alpha, \omega) > \theta$ we declare the document to be *relevant*, else we declare the document to be *nonrelevant*. As we know from Figure 14.8 (page 301), all points that satisfy $Score(\alpha, \omega) = \theta$ form a line (shown as a dashed line in Figure 15.7) and we thus have a linear classifier that separates relevant

from nonrelevant instances. Geometrically, we can find the separating line as follows. Consider the line passing through the plane $Score(\alpha, \omega)$ whose height is θ above the page containing Figure 15.7. Project this line down onto Figure 15.7; this will be the dashed line in Figure 15.7. Then, any subsequent query/document pair that falls below the dashed line in Figure 15.7 is deemed *nonrelevant*; above the dashed line, *relevant*.

Thus, the problem of making a binary *relevant/nonrelevant* judgment given training examples as above turns into one of learning the dashed line in Figure 15.7 separating *relevant* training examples from the *nonrelevant* ones. Being in the $\alpha\text{-}\omega$ plane, this line can be written as a linear equation involving α and ω , with two parameters (slope and intercept). The methods of linear classification that we have already looked at in Chapters 13–15 provide methods for choosing this line. Provided we can build a sufficiently rich collection of training samples, we can thus altogether avoid hand-tuning score functions as in Section 7.2.3 (page 145). The bottleneck of course is the ability to maintain a suitably representative set of training examples, whose relevance assessments must be made by experts.

15.4.2 Result ranking by machine learning

The above ideas can be readily generalized to functions of many more than two variables. There are lots of other scores that are indicative of the relevance of a document to a query, including static quality (PageRank-style measures, discussed in Chapter 21), document age, zone contributions, document length, and so on. Providing that these measures can be calculated for a training document collection with relevance judgments, any number of such measures can be used to train a machine learning classifier. For instance, we could train an SVM over binary relevance judgments, and order documents based on their probability of relevance, which is monotonic with the documents' signed distance from the decision boundary.

However, approaching IR result ranking like this is not necessarily the right way to think about the problem. Statisticians normally first divide problems into classification problems (where a categorical variable is predicted) versus *regression* problems (where a real number is predicted). In between is the specialized field of *ordinal regression* where a ranking is predicted. Machine learning for ad hoc retrieval is most properly thought of as an ordinal regression problem, where the goal is to rank a set of documents for a query, given training data of the same sort. This formulation gives some additional power, since documents can be evaluated relative to other candidate documents for the same query, rather than having to be mapped to a global scale of goodness, while also weakening the problem space, since just a ranking is required rather than an absolute measure of relevance. Issues of ranking are especially germane in web search, where the ranking at

the very top of the results list is exceedingly important, whereas decisions of relevance of a document to a query may be much less important. Such work can and has been pursued using the structural SVM framework which we mentioned in Section 15.2.2, where the class being predicted is a ranking of results for a query, but here we will present the slightly simpler ranking SVM.

RANKING SVM

The construction of a *ranking SVM* proceeds as follows. We begin with a set of judged queries. For each training query q , we have a set of documents returned in response to the query, which have been totally ordered by a person for relevance to the query. We construct a vector of features $\psi_j = \psi(d_j, q)$ for each document/query pair, using features such as those discussed in Section 15.4.1, and many more. For two documents d_i and d_j , we then form the vector of feature differences:

$$(15.18) \quad \Phi(d_i, d_j, q) = \psi(d_i, q) - \psi(d_j, q)$$

By hypothesis, one of d_i and d_j has been judged more relevant. If d_i is judged more relevant than d_j , denoted $d_i \prec d_j$ (d_i should precede d_j in the results ordering), then we will assign the vector $\Phi(d_i, d_j, q)$ the class $y_{ijq} = +1$; otherwise -1 . The goal then is to build a classifier which will return

$$(15.19) \quad \vec{w}^T \Phi(d_i, d_j, q) > 0 \quad \text{iff} \quad d_i \prec d_j$$

This SVM learning task is formalized in a manner much like the other examples that we saw before:

$$(15.20) \quad \text{Find } \vec{w}, \text{ and } \xi_{i,j} \geq 0 \text{ such that:}$$

- $\frac{1}{2} \vec{w}^T \vec{w} + C \sum_{i,j} \xi_{i,j}$ is minimized
- and for all $\{\Phi(d_i, d_j, q) : d_i \prec d_j\}$, $\vec{w}^T \Phi(d_i, d_j, q) \geq 1 - \xi_{i,j}$

We can leave out y_{ijq} in the statement of the constraint, since we only need to consider the constraint for document pairs ordered in one direction, since \prec is antisymmetric. These constraints are then solved, as before, to give a linear classifier which can rank pairs of documents. This approach has been used to build ranking functions which outperform standard hand-built ranking functions in IR evaluations on standard data sets; see the references for papers that present such results.

Both of the methods that we have just looked at use a linear weighting of document features that are indicators of relevance, as has most work in this area. It is therefore perhaps interesting to note that much of traditional IR weighting involves *nonlinear* scaling of basic measurements (such as log-weighting of term frequency, or idf). At the present time, machine learning is very good at producing optimal weights for features in a linear combination

(or other similar restricted model classes), but it is not good at coming up with good nonlinear scalings of basic measurements. This area remains the domain of human feature engineering.

The idea of learning ranking functions has been around for a number of years, but it is only very recently that sufficient machine learning knowledge, training document collections, and computational power have come together to make this method practical and exciting. It is thus too early to write something definitive on machine learning approaches to ranking in information retrieval, but there is every reason to expect the use and importance of machine learned ranking approaches to grow over time. While skilled humans can do a very good job at defining ranking functions by hand, hand tuning is difficult, and it has to be done again for each new document collection and class of users.



Exercise 15.7

Plot the first 7 rows of Table 15.3 in the $\alpha\text{-}\omega$ plane to produce a figure like that in Figure 15.7.

Exercise 15.8

Write down the equation of a line in the $\alpha\text{-}\omega$ plane separating the R's from the Ns.

Exercise 15.9

Give a training example (consisting of values for α, ω and the relevance judgment) that when added to the training set makes it impossible to separate the R's from the N's using a line in the $\alpha\text{-}\omega$ plane.

15.5 References and further reading

The somewhat quirky name support vector machine originates in the neural networks literature, where learning algorithms were thought of as architectures, and often referred to as “machines”. The distinctive element of this model is that the decision boundary to use is completely decided (“supported”) by a few training data points, the support vectors.

For a more detailed presentation of SVMs, a good, well-known article-length introduction is (Burges 1998). Chen et al. (2005) introduce the more recent ν -SVM, which provides an alternative parameterization for dealing with inseparable problems, whereby rather than specifying a penalty C , you specify a parameter ν which bounds the number of examples which can appear on the wrong side of the decision surface. There are now also several books dedicated to SVMs, large margin learning, and kernels: (Cristianini and Shawe-Taylor 2000) and (Schölkopf and Smola 2001) are more mathematically oriented, while (Shawe-Taylor and Cristianini 2004) aims to be more practical. For the foundations by their originator, see (Vapnik 1998).

Some recent, more general books on statistical learning, such as (Hastie et al. 2001) also give thorough coverage of SVMs.

The construction of *multiclass SVMs* is discussed in (Weston and Watkins 1999), (Crammer and Singer 2001), and (Tsochantaridis et al. 2005). The last reference provides an introduction to the general framework of structural SVMs.

The kernel trick was first presented in (Aizerman et al. 1964). For more about string kernels and other kernels for structured data, see (Lodhi et al. 2002) and (Gaertner et al. 2002). The Advances in Neural Information Processing (NIPS) conferences have become the premier venue for theoretical machine learning work, such as on SVMs. Other venues such as SIGIR are much stronger on experimental methodology and using text-specific features to improve classifier effectiveness.

A recent comparison of most current machine learning classifiers (though on problems rather different from typical text problems) can be found in (Caruana and Niculescu-Mizil 2006). (Li and Yang 2003), discussed in Section 13.6, is the most recent comparative evaluation of machine learning classifiers on text classification. Older examinations of classifiers on text problems can be found in (Yang 1999, Yang and Liu 1999, Dumais et al. 1998). Joachims (2002a) presents his work on SVMs applied to text problems in detail. Zhang and Oles (2001) present an insightful comparison of Naive Bayes, regularized logistic regression and SVM classifiers.

Joachims (1999) discusses methods of making SVM learning practical over large text data sets. Joachims (2006a) improves on this work.

A number of approaches to hierarchical classification have been developed in order to deal with the common situation where the classes to be assigned have a natural hierarchical organization (Koller and Sahami 1997, McCallum et al. 1998, Weigend et al. 1999, Dumais and Chen 2000). In a recent large study on scaling SVMs to the entire Yahoo! directory, Liu et al. (2005) conclude that hierarchical classification noticeably if still modestly outperforms flat classification. Classifier effectiveness remains limited by the very small number of training documents for many classes. For a more general approach that can be applied to modeling relations between classes, which may be arbitrary rather than simply the case of a hierarchy, see Tsochantaridis et al. (2005).

Moschitti and Basili (2004) investigate the use of complex nominals, proper nouns and word senses as features in text classification.

Dietterich (2002) overviews ensemble methods for classifier combination, while Schapire (2003) focuses particularly on boosting, which is applied to text classification in (Schapire and Singer 2000).

Chapelle et al. (2006) present an introduction to work in semi-supervised methods, including in particular chapters on using EM for semi-supervised text classification (Nigam et al. 2006) and on transductive SVMs (Joachims

2006b). Sindhwan and Keerthi (2006) present a more efficient implementation of a transductive SVM for large data sets.

Tong and Koller (2001) explore active learning with SVMs for text classification; Baldridge and Osborne (2004) point out that examples selected for annotation with one classifier in an active learning context may be no better than random examples when used with another classifier.

Machine learning approaches to ranking for ad hoc retrieval were pioneered in (Wong et al. 1988), (Fuhr 1992), and (Gey 1994). But limited training data and poor machine learning techniques meant that these pieces of work achieved only middling results, and hence they only had limited impact at the time.

Taylor et al. (2006) study using machine learning to tune the parameters of the BM25 family of ranking functions (Section 11.4.3, page 232) so as to maximize NDCG (Section 8.4, page 163). Machine learning approaches to ordinal regression appear in (Herbrich et al. 2000) and (Burges et al. 2005), and are applied to clickstream data in (Joachims 2002b). Cao et al. (2006) study how to make this approach effective in IR, and Qin et al. (2007) suggest an extension involving using multiple hyperplanes. Yue et al. (2007) study how to do ranking with a structural SVM approach, and in particular show how this construction can be effectively used to directly optimize for MAP (Section 8.4, page 158), rather than using surrogate measures like accuracy or area under the ROC curve. Geng et al. (2007) study feature selection for the ranking problem.

Other approaches to learning to rank have also been shown to be effective for web search, such as (Burges et al. 2005, Richardson et al. 2006).

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19

Web search basics

In this and the following two chapters, we consider web search engines. Sections 19.1–19.4 provide some background and history to help the reader appreciate the forces that conspire to make the Web chaotic, fast-changing and (from the standpoint of information retrieval) very different from the “traditional” collections studied thus far in this book. Sections 19.5–19.6 deal with estimating the number of documents indexed by web search engines, and the elimination of duplicate documents in web indexes, respectively. These two latter sections serve as background material for the following two chapters.

19.1 Background and history

The Web is unprecedented in many ways: unprecedented in scale, unprecedented in the almost-complete lack of coordination in its creation, and unprecedented in the diversity of backgrounds and motives of its participants. Each of these contributes to making web search different – and generally far harder – than searching “traditional” documents.

The invention of hypertext, envisioned by Vannevar Bush in the 1940’s and first realized in working systems in the 1970’s, significantly precedes the formation of the World Wide Web (which we will simply refer to as the Web), in the 1990’s. Web usage has shown tremendous growth to the point where it now claims a good fraction of humanity as participants, by relying on a simple, open client-server design: (1) the server communicates with the client via a protocol (the *http* or hypertext transfer protocol) that is lightweight and simple, asynchronously carrying a variety of payloads (text, images and – over time – richer media such as audio and video files) encoded in a simple markup language called *HTML* (for hypertext markup language); (2) the client – generally a *browser*, an application within a graphical user environment – can ignore what it does not understand. Each of these seemingly innocuous features has contributed enormously to the growth of the Web, so it is worthwhile to examine them further.

HTTP

HTML

The basic operation is as follows: a client (such as a browser) sends an *http request* to a *web server*. The browser specifies a *URL* (for *Universal Resource Locator*) such as `http://www.stanford.edu/home/atoz/contact.html`. In this example URL, the string `http` refers to the protocol to be used for transmitting the data. The string `www.stanford.edu` is known as the *domain* and specifies the root of a hierarchy of web pages (typically mirroring a filesystem hierarchy underlying the web server). In this example, `/home/atoz/contact.html` is a path in this hierarchy with a file `contact.html` that contains the information to be returned by the web server at `www.stanford.edu` in response to this request. The HTML-encoded file `contact.html` holds the hyperlinks and the content (in this instance, contact information for Stanford University), as well as formatting rules for rendering this content in a browser. Such an http request thus allows us to fetch the content of a page, something that will prove to be useful to us for crawling and indexing documents (Chapter 20).

The designers of the first browsers made it easy to view the HTML markup tags on the content of a URL. This simple convenience allowed new users to create their own HTML content without extensive training or experience; rather, they learned from example content that they liked. As they did so, a second feature of browsers supported the rapid proliferation of web content creation and usage: browsers ignored what they did not understand. This did not, as one might fear, lead to the creation of numerous incompatible dialects of HTML. What it did promote was amateur content creators who could freely experiment with and learn from their newly created web pages without fear that a simple syntax error would “bring the system down.” Publishing on the Web became a mass activity that was not limited to a few trained programmers, but rather open to tens and eventually hundreds of millions of individuals. For most users and for most information needs, the Web quickly became the best way to supply and consume information on everything from rare ailments to subway schedules.

The mass publishing of information on the Web is essentially useless unless this wealth of information can be discovered and consumed by other users. Early attempts at making web information “discoverable” fell into two broad categories: (1) full-text index search engines such as Altavista, Excite and Infoseek and (2) taxonomies populated with web pages in categories, such as Yahoo! The former presented the user with a keyword search interface supported by inverted indexes and ranking mechanisms building on those introduced in earlier chapters. The latter allowed the user to browse through a hierarchical tree of category labels. While this is at first blush a convenient and intuitive metaphor for finding web pages, it has a number of drawbacks: first, accurately classifying web pages into taxonomy tree nodes is for the most part a manual editorial process, which is difficult to scale with the size of the Web. Arguably, we only need to have “high-quality”

web pages in the taxonomy, with only the best web pages for each category. However, just discovering these and classifying them accurately and consistently into the taxonomy entails significant human effort. Furthermore, in order for a user to effectively discover web pages classified into the nodes of the taxonomy tree, the user’s idea of what sub-tree(s) to seek for a particular topic should match that of the editors performing the classification. This quickly becomes challenging as the size of the taxonomy grows; the Yahoo! taxonomy tree surpassed 1000 distinct nodes fairly early on. Given these challenges, the popularity of taxonomies declined over time, even though variants (such as About.com and the Open Directory Project) sprang up with subject-matter experts collecting and annotating web pages for each category.

The first generation of web search engines transported classical search techniques such as those in the preceding chapters to the web domain, focusing on the challenge of scale. The earliest web search engines had to contend with indexes containing tens of millions of documents, which was a few orders of magnitude larger than any prior information retrieval system in the public domain. Indexing, query serving and ranking at this scale required the harnessing together of tens of machines to create highly available systems, again at scales not witnessed hitherto in a consumer-facing search application. The first generation of web search engines was largely successful at solving these challenges while continually indexing a significant fraction of the Web, all the while serving queries with sub-second response times. However, the quality and relevance of web search results left much to be desired owing to the idiosyncrasies of content creation on the Web that we discuss in Section 19.2. This necessitated the invention of new ranking and spam-fighting techniques in order to ensure the quality of the search results. While classical information retrieval techniques (such as those covered earlier in this book) continue to be necessary for web search, they are not by any means sufficient. A key aspect (developed further in Chapter 21) is that whereas classical techniques measure the relevance of a document to a query, there remains a need to gauge the *authoritativeness* of a document based on cues such as which website hosts it.

19.2 Web characteristics

The essential feature that led to the explosive growth of the web – decentralized content publishing with essentially no central control of authorship – turned out to be the biggest challenge for web search engines in their quest to index and retrieve this content. Web page authors created content in dozens of (natural) languages and thousands of dialects, thus demanding many different forms of stemming and other linguistic operations. Because publish-

ing was now open to tens of millions, web pages exhibited heterogeneity at a daunting scale, in many crucial aspects. First, content-creation was no longer the privy of editorially-trained writers; while this represented a tremendous democratization of content creation, it also resulted in a tremendous variation in grammar and style (and in many cases, no recognizable grammar or style). Indeed, web publishing in a sense unleashed the best and worst of desktop publishing on a planetary scale, so that pages quickly became riddled with wild variations in colors, fonts and structure. Some web pages, including the professionally created home pages of some large corporations, consisted entirely of images (which, when clicked, led to richer textual content) – and therefore, no indexable text.

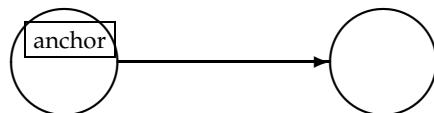
What about the substance of the text in web pages? The democratization of content creation on the web meant a new level of granularity in *opinion* on virtually any subject. This meant that the web contained truth, lies, contradictions and suppositions on a grand scale. This gives rise to the question: which web pages does one trust? In a simplistic approach, one might argue that some publishers are trustworthy and others not – begging the question of how a search engine is to assign such a measure of trust to each website or web page. In Chapter 21 we will examine approaches to understanding this question. More subtly, there may be no universal, user-independent notion of trust; a web page whose contents are trustworthy to one user may not be so to another. In traditional (non-web) publishing this is not an issue: users self-select sources they find trustworthy. Thus one reader may find the reporting of *The New York Times* to be reliable, while another may prefer *The Wall Street Journal*. But when a search engine is the only viable means for a user to become aware of (let alone select) most content, this challenge becomes significant.

While the question “how big is the Web?” has no easy answer (see Section 19.5), the question “how many web pages are in a search engine’s index” is more precise, although, even this question has issues. By the end of 1995, Altavista reported that it had crawled and indexed approximately 30 million *static web pages*. Static web pages are those whose content does not vary from one request for that page to the next. For this purpose, a professor who manually updates his home page every week is considered to have a static web page, but an airport’s flight status page is considered to be dynamic. Dynamic pages are typically mechanically generated by an application server in response to a query to a database, as shown in Figure 19.1. One sign of such a page is that the URL has the character “?” in it. Since the number of static web pages was believed to be doubling every few months in 1995, early web search engines such as Altavista had to constantly add hardware and bandwidth for crawling and indexing web pages.

STATIC WEB PAGES



► **Figure 19.1** A dynamically generated web page. The browser sends a request for flight information on flight AA129 to the web application, that fetches the information from back-end databases then creates a dynamic web page that it returns to the browser.



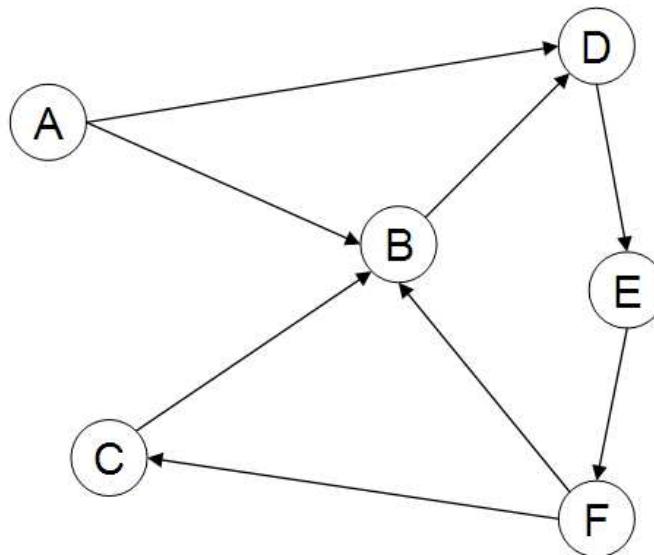
► **Figure 19.2** Two nodes of the web graph joined by a link.

19.2.1 The web graph

We can view the static Web consisting of static HTML pages together with the hyperlinks between them as a directed graph in which each web page is a node and each hyperlink a directed edge.

ANCHOR TEXT
IN-LINKS
OUT-LINKS

Figure 19.2 shows two nodes A and B from the web graph, each corresponding to a web page, with a hyperlink from A to B. We refer to the set of all such nodes and directed edges as the web graph. Figure 19.2 also shows that (as is the case with most links on web pages) there is some text surrounding the origin of the hyperlink on page A. This text is generally encapsulated in the `href` attribute of the `<a>` (or anchor) tag that encodes the hyperlink in the HTML code of page A, and is referred to as *anchor text*. As one might suspect, this directed graph is not *strongly connected*: there are pairs of pages such that one cannot proceed from one page of the pair to the other by following hyperlinks. We refer to the hyperlinks into a page as *in-links* and those out of a page as *out-links*. The number of in-links to a page (also known as its *in-degree*) has averaged from roughly 8 to 15, in a range of studies. We similarly define the out-degree of a web page to be the number of links out



► **Figure 19.3** A sample small web graph. In this example we have six pages labeled A-F. Page B has in-degree 3 and out-degree 1. This example graph is not strongly connected: there is no path from any of pages B-F to page A.

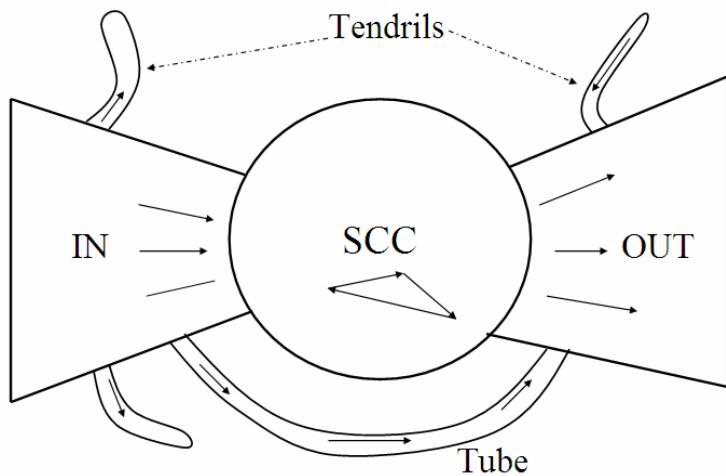
of it. These notions are represented in Figure 19.3.

POWER LAW

BOWTIE

There is ample evidence that these links are not randomly distributed; for one thing, the distribution of the number of links into a web page does not follow the Poisson distribution one would expect if every web page were to pick the destinations of its links uniformly at random. Rather, this distribution is widely reported to be a *power law*, in which the total number of web pages with in-degree i is proportional to $1/i^\alpha$; the value of α typically reported by studies is 2.1.¹ Furthermore, several studies have suggested that the directed graph connecting web pages has a *bowtie* shape: there are three major categories of web pages that are sometimes referred to as IN, OUT and SCC. A web surfer can pass from any page in IN to any page in SCC, by following hyperlinks. Likewise, a surfer can pass from page in SCC to any page in OUT. Finally, the surfer can surf from any page in SCC to any other page in SCC. However, it is not possible to pass from a page in SCC to any page in IN, or from a page in OUT to a page in SCC (or, consequently, IN). Notably, in several studies IN and OUT are roughly equal in size, whereas

¹. Cf. Zipf's law of the distribution of words in text in Chapter 5 (page 90), which is a power law with $\alpha = 1$.



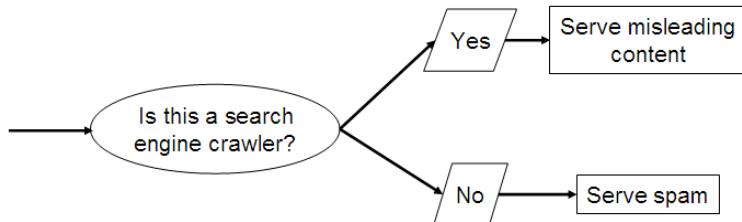
► **Figure 19.4** The bowtie structure of the Web. Here we show one tube and three tendrils.

SCC is somewhat larger; most web pages fall into one of these three sets. The remaining pages form into *tubes* that are small sets of pages outside SCC that lead directly from IN to OUT, and *tendrils* that either lead nowhere from IN, or from nowhere to OUT. Figure 19.4 illustrates this structure of the Web.

19.2.2 Spam

Early in the history of web search, it became clear that web search engines were an important means for connecting advertisers to prospective buyers. A user searching for maui golf real estate is not merely seeking news or entertainment on the subject of housing on golf courses on the island of Maui, but instead likely to be seeking to purchase such a property. Sellers of such property and their agents, therefore, have a strong incentive to create web pages that rank highly on this query. In a search engine whose scoring was based on term frequencies, a web page with numerous repetitions of maui golf real estate would rank highly. This led to the first generation of *spam*, which (in the context of web search) is the manipulation of web page content for the purpose of appearing high up in search results for selected keywords. To avoid irritating users with these repetitions, sophisticated *spammers* resorted to such tricks as rendering these repeated terms in the same color as the background. Despite these words being consequently invisible to the human user, a search engine indexer would parse the invisible words out of

SPAM



► **Figure 19.5** Cloaking as used by spammers.

the HTML representation of the web page and index these words as being present in the page.

At its root, spam stems from the heterogeneity of motives in content creation on the Web. In particular, many web content creators have commercial motives and therefore stand to gain from manipulating search engine results. You might argue that this is no different from a company that uses large fonts to list its phone numbers in the yellow pages; but this generally costs the company more and is thus a fairer mechanism. A more apt analogy, perhaps, is the use of company names beginning with a long string of A's to be listed early in a yellow pages category. In fact, the yellow pages' model of companies paying for larger/darker fonts has been replicated in web search: in many search engines, it is possible to pay to have one's web page included in the search engine's index – a model known as *paid inclusion*. Different search engines have different policies on whether to allow paid inclusion, and whether such a payment has any effect on ranking in search results.

Search engines soon became sophisticated enough in their spam detection to screen out a large number of repetitions of particular keywords. Spammers responded with a richer set of spam techniques, the best known of which we now describe. The first of these techniques is *cloaking*, shown in Figure 19.5. Here, the spammer's web server returns different pages depending on whether the http request comes from a web search engine's crawler (the part of the search engine that gathers web pages, to be described in Chapter 20), or from a human user's browser. The former causes the web page to be indexed by the search engine under misleading keywords. When the user searches for these keywords and elects to view the page, he receives a web page that has altogether different content than that indexed by the search engine. Such deception of search indexers is unknown in the traditional world of information retrieval; it stems from the fact that the relationship between page publishers and web search engines is not completely collaborative.

A *doorway page* contains text and metadata carefully chosen to rank highly



on selected search keywords. When a browser requests the doorway page, it is redirected to a page containing content of a more commercial nature. More complex spamming techniques involve manipulation of the metadata related to a page including (for reasons we will see in Chapter 21) the links into a web page. Given that spamming is inherently an economically motivated activity, there has sprung around it an industry of *Search Engine Optimizers*, or SEOs to provide consultancy services for clients who seek to have their web pages rank highly on selected keywords. Web search engines frown on this business of attempting to decipher and adapt to their proprietary ranking techniques and indeed announce policies on forms of SEO behavior they do not tolerate (and have been known to shut down search requests from certain SEOs for violation of these). Inevitably, the parrying between such SEOs (who gradually infer features of each web search engine's ranking methods) and the web search engines (who adapt in response) is an unending struggle; indeed, the research sub-area of *adversarial information retrieval* has sprung up around this battle. To combat spammers who manipulate the text of their web pages is the exploitation of the link structure of the Web – a technique known as *link analysis*. The first web search engine known to apply link analysis on a large scale (to be detailed in Chapter 21) was Google, although all web search engines currently make use of it (and correspondingly, spammers now invest considerable effort in subverting it – this is known as *link spam*).

Exercise 19.1

If the number of pages with in-degree i is proportional to $1/i^{2.1}$, what is the probability that a randomly chosen web page has in-degree 1?

Exercise 19.2

If the number of pages with in-degree i is proportional to $1/i^{2.1}$, what is the average in-degree of a web page?

Exercise 19.3

If the number of pages with in-degree i is proportional to $1/i^{2.1}$, then as the largest in-degree goes to infinity, does the fraction of pages with in-degree i grow, stay the same, or diminish? How would your answer change for values of the exponent other than 2.1?

Exercise 19.4

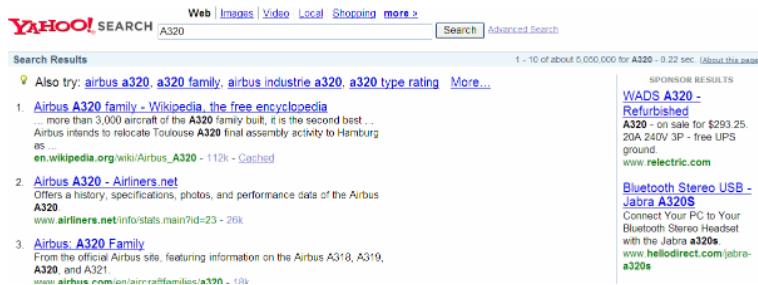
The average in-degree of all nodes in a snapshot of the web graph is 9. What can we say about the average out-degree of all nodes in this snapshot?

19.3 Advertising as the economic model

Early in the history of the Web, companies used graphical banner advertisements on web pages at popular websites (news and entertainment sites such as MSN, America Online, Yahoo! and CNN). The primary purpose of these advertisements was *branding*: to convey to the viewer a positive feeling about

CPM	the brand of the company placing the advertisement. Typically these advertisements are priced on a <i>cost per mil</i> (CPM) basis: the cost to the company of having its banner advertisement displayed 1000 times. Some websites struck contracts with their advertisers in which an advertisement was priced not by the number of times it is displayed (also known as <i>impressions</i>), but rather by the number of times it was <i>clicked on</i> by the user. This pricing model is known as the <i>cost per click</i> (CPC) model.
CPC	In such cases, clicking on the advertisement leads the user to a web page set up by the advertiser, where the user is induced to make a purchase. Here the goal of the advertisement is not so much brand promotion as to induce a transaction. This distinction between brand and transaction-oriented advertising was already widely recognized in the context of conventional media such as broadcast and print. The interactivity of the web allowed the CPC billing model – clicks could be metered and monitored by the website and billed to the advertiser.
SPONSORED SEARCH SEARCH ADVERTISING	The pioneer in this direction was a company named Goto, which changed its name to Overture prior to eventual acquisition by Yahoo! Goto was not, in the traditional sense, a search engine; rather, for every query term q it accepted <i>bids</i> from companies who wanted their web page shown on the query q . In response to the query q , Goto would return the pages of all advertisers who bid for q , ordered by their bids. Furthermore, when the user clicked on one of the returned results, the corresponding advertiser would make a payment to Goto (in the initial implementation, this payment equaled the advertiser's bid for q).
ALGORITHMIC SEARCH	Several aspects of Goto's model are worth highlighting. First, a user typing the query q into Goto's search interface was actively expressing an interest and intent related to the query q . For instance, a user typing golf clubs is more likely to be imminently purchasing a set than one who is simply browsing news on golf. Second, Goto only got compensated when a user actually expressed interest in an advertisement – as evinced by the user clicking the advertisement. Taken together, these created a powerful mechanism by which to connect advertisers to consumers, quickly raising the annual revenues of Goto/Overture into hundreds of millions of dollars. This style of search engine came to be known variously as <i>sponsored search</i> or <i>search advertising</i> .

Given these two kinds of search engines – the “pure” search engines such as Google and Altavista, versus the sponsored search engines – the logical next step was to combine them into a single user experience. Current search engines follow precisely this model: they provide pure search results (generally known as *algorithmic search* results) as the primary response to a user's search, together with sponsored search results displayed separately and distinctively to the right of the algorithmic results. This is shown in Figure 19.6. Retrieving sponsored search results and ranking them in response to a query has now become considerably more sophisticated than the simple Goto scheme; the process entails a blending of ideas from information



► **Figure 19.6** Search advertising triggered by query keywords. Here the query A320 returns algorithmic search results about the Airbus aircraft, together with advertisements for various non-aircraft goods numbered A320, that advertisers seek to market to those querying on this query. The lack of advertisements for the aircraft reflects the fact that few marketers attempt to sell A320 aircraft on the web.

SEARCH ENGINE
MARKETING

CLICK SPAM

retrieval and microeconomics, and is beyond the scope of this book. For advertisers, understanding how search engines do this ranking and how to allocate marketing campaign budgets to different keywords and to different sponsored search engines has become a profession known as *search engine marketing* (SEM).

The inherently economic motives underlying sponsored search give rise to attempts by some participants to subvert the system to their advantage. This can take many forms, one of which is known as *click spam*. There is currently no universally accepted definition of click spam. It refers (as the name suggests) to clicks on sponsored search results that are not from bona fide search users. For instance, a devious advertiser may attempt to exhaust the advertising budget of a competitor by clicking repeatedly (through the use of a robotic click generator) on that competitor's sponsored search advertisements. Search engines face the challenge of discerning which of the clicks they observe are part of a pattern of click spam, to avoid charging their advertiser clients for such clicks.

?

Exercise 19.5

The Goto method ranked advertisements matching a query by *bid*: the highest-bidding advertiser got the top position, the second-highest the next, and so on. What can go wrong with this when the highest-bidding advertiser places an advertisement that is irrelevant to the query? Why might an advertiser with an irrelevant advertisement bid high in this manner?

Exercise 19.6

Suppose that, in addition to bids, we had for each advertiser their *click-through rate*: the ratio of the historical number of times users click on their advertisement to the number of times the advertisement was shown. Suggest a modification of the Goto scheme that exploits this data to avoid the problem in Exercise 19.5 above.

19.4 The search user experience

It is crucial that we understand the users of web search as well. This is again a significant change from traditional information retrieval, where users were typically professionals with at least some training in the art of phrasing queries over a well-authored collection whose style and structure they understood well. In contrast, web search users tend to not know (or care) about the heterogeneity of web content, the syntax of query languages and the art of phrasing queries; indeed, a mainstream tool (as web search has come to become) should not place such onerous demands on billions of people. A range of studies has concluded that the average number of keywords in a web search is somewhere between 2 and 3. Syntax operators (Boolean connectives, wildcards, etc.) are seldom used, again a result of the composition of the audience – “normal” people, not information scientists.

It is clear that the more user traffic a web search engine can attract, the more revenue it stands to earn from sponsored search. How do search engines differentiate themselves and grow their traffic? Here Google identified two principles that helped it grow at the expense of its competitors: (1) a focus on relevance, specifically precision rather than recall in the first few results; (2) a user experience that is lightweight, meaning that both the search query page and the search results page are uncluttered and almost entirely textual, with very few graphical elements. The effect of the first was simply to save users time in locating the information they sought. The effect of the second is to provide a user experience that is extremely responsive, or at any rate not bottlenecked by the time to load the search query or results page.

19.4.1 User query needs

There appear to be three broad categories into which common web search queries can be grouped: (i) informational, (ii) navigational and (iii) transactional. We now explain these categories; it should be clear that some queries will fall in more than one of these categories, while others will fall outside them.

INFORMATIONAL QUERIES

Informational queries seek general information on a broad topic, such as leukemia or Provence. There is typically not a single web page that contains all the information sought; indeed, users with informational queries typically try to assimilate information from multiple web pages.

NAVIGATIONAL QUERIES

Navigational queries seek the website or home page of a single entity that the user has in mind, say Lufthansa airlines. In such cases, the user’s expectation is that the very first search result should be the home page of Lufthansa. The user is not interested in a plethora of documents containing the term Lufthansa; for such a user, the best measure of user satisfaction is precision at 1.

TRANSACTIONAL
QUERY

A *transactional query* is one that is a prelude to the user performing a transaction on the Web – such as purchasing a product, downloading a file or making a reservation. In such cases, the search engine should return results listing services that provide form interfaces for such transactions.

Discerning which of these categories a query falls into can be challenging. The category not only governs the algorithmic search results, but the suitability of the query for sponsored search results (since the query may reveal an intent to purchase). For navigational queries, some have argued that the search engine should return only a single result or even the target web page directly. Nevertheless, web search engines have historically engaged in a battle of bragging rights over which one indexes more web pages. Does the user really care? Perhaps not, but the media does highlight estimates (often statistically indefensible) of the sizes of various search engines. Users are influenced by these reports and thus, search engines do have to pay attention to how their index sizes compare to competitors'. For informational (and to a lesser extent, transactional) queries, the user does care about the comprehensiveness of the search engine.

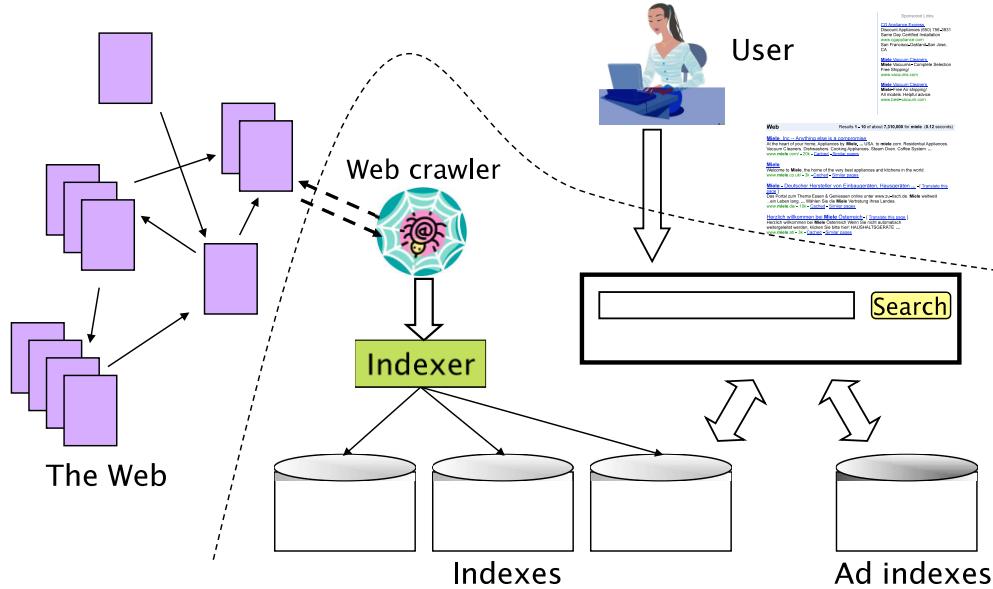
Figure 19.7 shows a composite picture of a web search engine including the crawler, as well as both the web page and advertisement indexes. The portion of the figure under the curved dashed line is internal to the search engine.

19.5 Index size and estimation

To a first approximation, comprehensiveness grows with index size, although it does matter which specific pages a search engine indexes – some pages are more informative than others. It is also difficult to reason about the fraction of the Web indexed by a search engine, because there is an infinite number of dynamic web pages; for instance, `http://www.yahoo.com/any_string` returns a valid HTML page rather than an error, politely informing the user that there is no such page at Yahoo! Such a "soft 404 error" is only one example of many ways in which web servers can generate an infinite number of valid web pages. Indeed, some of these are malicious spider traps devised to cause a search engine's crawler (the component that systematically gathers web pages for the search engine's index, described in Chapter 20) to stay within a spammer's website and index many pages from that site.

We could ask the following better-defined question: given two search engines, what are the relative sizes of their indexes? Even this question turns out to be imprecise, because:

1. In response to queries a search engine can return web pages whose contents it has not (fully or even partially) indexed. For one thing, search engines generally index only the first few thousand words in a web page.



► **Figure 19.7** The various components of a web search engine.

In some cases, a search engine is aware of a page p that is *linked to* by pages it has indexed, but has not indexed p itself. As we will see in Chapter 21, it is still possible to meaningfully return p in search results.

2. Search engines generally organize their indexes in various tiers and partitions, not all of which are examined on every search (recall tiered indexes from Section 7.2.1). For instance, a web page deep inside a website may be indexed but not retrieved on general web searches; it is however retrieved as a result of a search that a user has explicitly restricted to that website (such site-specific search is offered by most web search engines).

Thus, search engine indexes include multiple classes of indexed pages, so that there is no single measure of index size. These issues notwithstanding, a number of techniques have been devised for crude estimates of the ratio of the index sizes of two search engines, E_1 and E_2 . The basic hypothesis underlying these techniques is that each search engine indexes a fraction of the Web chosen independently and uniformly at random. This involves some questionable assumptions: first, that there is a finite size for the Web from which each search engine chooses a subset, and second, that each engine chooses an independent, uniformly chosen subset. As will be clear from the discussion of crawling in Chapter 20, this is far from true. However, if we begin

with these assumptions, then we can invoke a classical estimation technique known as the *capture-recapture method*.

Suppose that we could pick a random page from the index of E_1 and test whether it is in E_2 's index and symmetrically, test whether a random page from E_2 is in E_1 . These experiments give us fractions x and y such that our estimate is that a fraction x of the pages in E_1 are in E_2 , while a fraction y of the pages in E_2 are in E_1 . Then, letting $|E_i|$ denote the size of the index of search engine E_i , we have

$$x|E_1| \approx y|E_2|,$$

from which we have the form we will use

$$(19.1) \quad \frac{|E_1|}{|E_2|} \approx \frac{y}{x}.$$

If our assumption about E_1 and E_2 being independent and uniform random subsets of the Web were true, and our sampling process unbiased, then Equation (19.1) should give us an unbiased estimator for $|E_1|/|E_2|$. We distinguish between two scenarios here. Either the measurement is performed by someone with access to the index of one of the search engines (say an employee of E_1), or the measurement is performed by an independent party with no access to the innards of either search engine. In the former case, we can simply pick a random document from one index. The latter case is more challenging; by picking a random page from one search engine *from outside the search engine*, then verify whether the random page is present in the other search engine.

To implement the sampling phase, we might generate a random page from the entire (idealized, finite) Web and test it for presence in each search engine. Unfortunately, picking a web page uniformly at random is a difficult problem. We briefly outline several attempts to achieve such a sample, pointing out the biases inherent to each; following this we describe in some detail one technique that much research has built on.

1. *Random searches:* Begin with a search log of web searches; send a random search from this log to E_1 and a random page from the results. Since such logs are not widely available outside a search engine, one implementation is to trap all search queries going out of a work group (say scientists in a research center) that agrees to have all its searches logged. This approach has a number of issues, including the bias from the types of searches made by the work group. Further, a random document from the results of such a random search to E_1 is not the same as a random document from E_1 .
2. *Random IP addresses:* A second approach is to generate random IP addresses and send a request to a web server residing at the random address, collecting all pages at that server. The biases here include the fact

that many hosts might share one IP (due to a practice known as virtual hosting) or not accept http requests from the host where the experiment is conducted. Furthermore, this technique is more likely to hit one of the many sites with few pages, skewing the document probabilities; we may be able to correct for this effect if we understand the distribution of the number of pages on websites.

3. *Random walks:* If the web graph were a strongly connected directed graph, we could run a random walk starting at an arbitrary web page. This walk would converge to a steady state distribution (see Chapter 21, Section 21.2.1 for more background material on this), from which we could in principle pick a web page with a fixed probability. This method, too has a number of biases. First, the Web is not strongly connected so that, even with various corrective rules, it is difficult to argue that we can reach a steady state distribution starting from any page. Second, the time it takes for the random walk to settle into this steady state is unknown and could exceed the length of the experiment.

Clearly each of these approaches is far from perfect. We now describe a fourth sampling approach, *random queries*. This approach is noteworthy for two reasons: it has been successfully built upon for a series of increasingly refined estimates, and conversely it has turned out to be the approach most likely to be misinterpreted and carelessly implemented, leading to misleading measurements. The idea is to pick a page (almost) uniformly at random from a search engine's index by posing a random query to it. It should be clear that picking a set of random terms from (say) Webster's dictionary is not a good way of implementing this idea. For one thing, not all vocabulary terms occur equally often, so this approach will not result in documents being chosen uniformly at random from the search engine. For another, there are a great many terms in web documents that do not occur in a standard dictionary such as Webster's. To address the problem of vocabulary terms not in a standard dictionary, we begin by amassing a sample web dictionary. This could be done by crawling a limited portion of the Web, or by crawling a manually-assembled representative subset of the Web such as Yahoo! (as was done in the earliest experiments with this method). Consider a conjunctive query with two or more randomly chosen words from this dictionary.

Operationally, we proceed as follows: we use a random conjunctive query on E_1 and pick from the top 100 returned results a page p at random. We then test p for presence in E_2 by choosing 6-8 low-frequency terms in p and using them in a conjunctive query for E_2 . We can improve the estimate by repeating the experiment a large number of times. Both the sampling process and the testing process have a number of issues.

1. Our sample is biased towards longer documents.

2. Picking from the top 100 results of E_1 induces a bias from the ranking algorithm of E_1 . Picking from all the results of E_1 makes the experiment slower. This is particularly so because most web search engines put up defenses against excessive robotic querying.
3. During the checking phase, a number of additional biases are introduced: for instance, E_2 may not handle 8-word conjunctive queries properly.
4. Either E_1 or E_2 may refuse to respond to the test queries, treating them as robotic spam rather than as bona fide queries.
5. There could be operational problems like connection time-outs.

A sequence of research has built on this basic paradigm to eliminate some of these issues; there is no perfect solution yet, but the level of sophistication in statistics for understanding the biases is increasing. The main idea is to address biases by estimating, for each document, the magnitude of the bias. From this, standard statistical sampling methods can generate unbiased samples. In the checking phase, the newer work moves away from conjunctive queries to phrase and other queries that appear to be better-behaved. Finally, newer experiments use other sampling methods besides random queries. The best known of these is *document random walk sampling*, in which a document is chosen by a random walk on a virtual graph derived from documents. In this graph, nodes are documents; two documents are connected by an edge if they share two or more words in common. The graph is never instantiated; rather, a random walk on it can be performed by moving from a document d to another by picking a pair of keywords in d , running a query on a search engine and picking a random document from the results. Details may be found in the references in Section 19.7.

**Exercise 19.7**

Two web search engines A and B each generate a large number of pages uniformly at random from their indexes. 30% of A's pages are present in B's index, while 50% of B's pages are present in A's index. What is the number of pages in A's index relative to B's?

19.6 Near-duplicates and shingling

One aspect we have ignored in the discussion of index size in Section 19.5 is *duplication*: the Web contains multiple copies of the same content. By some estimates, as many as 40% of the pages on the Web are duplicates of other pages. Many of these are legitimate copies; for instance, certain information repositories are mirrored simply to provide redundancy and access reliability. Search engines try to avoid indexing multiple copies of the same content, to keep down storage and processing overheads.

The simplest approach to detecting duplicates is to compute, for each web page, a *fingerprint* that is a succinct (say 64-bit) digest of the characters on that page. Then, whenever the fingerprints of two web pages are equal, we test whether the pages themselves are equal and if so declare one of them to be a duplicate copy of the other. This simplistic approach fails to capture a crucial and widespread phenomenon on the Web: *near duplication*. In many cases, the contents of one web page are identical to those of another except for a few characters – say, a notation showing the date and time at which the page was last modified. Even in such cases, we want to be able to declare the two pages to be close enough that we only index one copy. Short of exhaustively comparing all pairs of web pages, an infeasible task at the scale of billions of pages, how can we detect and filter out such near duplicates?

SHINGLING

We now describe a solution to the problem of detecting near-duplicate web pages. The answer lies in a technique known as *shingling*. Given a positive integer k and a sequence of terms in a document d , define the k -shingles of d to be the set of all consecutive sequences of k terms in d . As an example, consider the following text: a rose is a rose is a rose. The 4-shingles for this text ($k = 4$ is a typical value used in the detection of near-duplicate web pages) are a rose is a, rose is a rose and is a rose is. The first two of these shingles each occur twice in the text. Intuitively, two documents are near duplicates if the sets of shingles generated from them are nearly the same. We now make this intuition precise, then develop a method for efficiently computing and comparing the sets of shingles for all web pages.

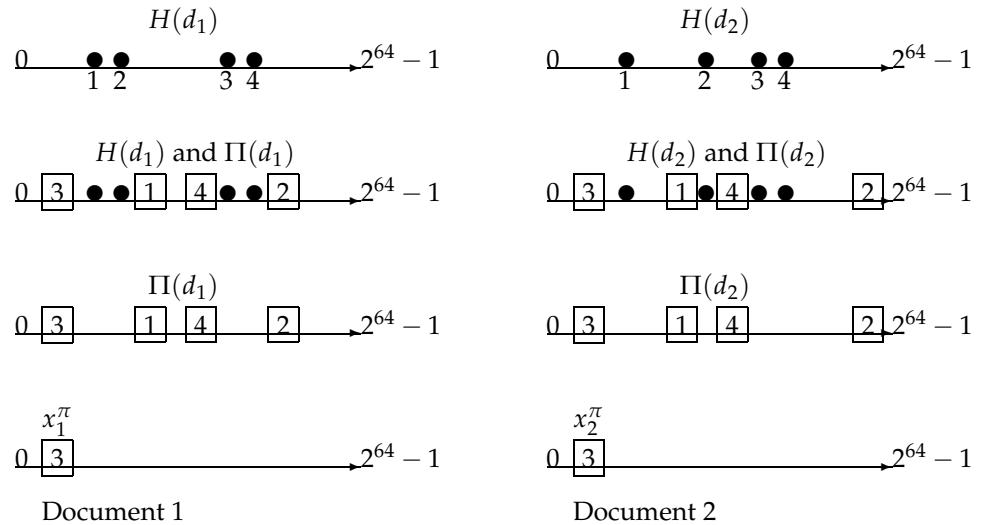
Let $S(d_j)$ denote the set of shingles of document d_j . Recall the Jaccard coefficient from page 61, which measures the degree of overlap between the sets $S(d_1)$ and $S(d_2)$ as $|S(d_1) \cap S(d_2)| / |S(d_1) \cup S(d_2)|$; denote this by $J(S(d_1), S(d_2))$. Our test for near duplication between d_1 and d_2 is to compute this Jaccard coefficient; if it exceeds a preset threshold (say, 0.9), we declare them near duplicates and eliminate one from indexing. However, this does not appear to have simplified matters: we still have to compute Jaccard coefficients pairwise.

To avoid this, we use a form of hashing. First, we map every shingle into a hash value over a large space, say 64 bits. For $j = 1, 2$, let $H(d_j)$ be the corresponding set of 64-bit hash values derived from $S(d_j)$. We now invoke the following trick to detect document pairs whose sets $H()$ have large Jaccard overlaps. Let π be a random permutation from the 64-bit integers to the 64-bit integers. Denote by $\Pi(d_j)$ the set of permuted hash values in $H(d_j)$; thus for each $h \in H(d_j)$, there is a corresponding value $\pi(h) \in \Pi(d_j)$.

Let x_j^π be the smallest integer in $\Pi(d_j)$. Then

Theorem 19.1.

$$J(S(d_1), S(d_2)) = P(x_1^\pi = x_2^\pi).$$



► **Figure 19.8** Illustration of shingle sketches. We see two documents going through four stages of shingle sketch computation. In the first step (top row), we apply a 64-bit hash to each shingle from each document to obtain $H(d_1)$ and $H(d_2)$ (circles). Next, we apply a random permutation Π to permute $H(d_1)$ and $H(d_2)$, obtaining $\Pi(d_1)$ and $\Pi(d_2)$ (squares). The third row shows only $\Pi(d_1)$ and $\Pi(d_2)$, while the bottom row shows the minimum values x_1^π and x_2^π for each document.

Proof. We give the proof in a slightly more general setting: consider a family of sets whose elements are drawn from a common universe. View the sets as columns of a matrix A , with one row for each element in the universe. The element $a_{ij} = 1$ if element i is present in the set S_j that the j th column represents.

Let Π be a random permutation of the rows of A ; denote by $\Pi(S_j)$ the column that results from applying Π to the j th column. Finally, let x_j^π be the index of the first row in which the column $\Pi(S_j)$ has a 1. We then prove that for any two columns j_1, j_2 ,

$$P(x_{j_1}^\pi = x_{j_2}^\pi) = J(S_{j_1}, S_{j_2}).$$

If we can prove this, the theorem follows.

Consider two columns j_1, j_2 as shown in Figure 19.9. The ordered pairs of entries of S_{j_1} and S_{j_2} partition the rows into four types: those with 0's in both of these columns, those with a 0 in S_{j_1} and a 1 in S_{j_2} , those with a 1 in S_{j_1} and a 0 in S_{j_2} , and finally those with 1's in both of these columns. Indeed, the first four rows of Figure 19.9 exemplify all of these four types of rows.

S_{j_1}	S_{j_2}
0	1
1	0
1	1
0	0
1	1
0	1

► **Figure 19.9** Two sets S_{j_1} and S_{j_2} ; their Jaccard coefficient is 2/5.

Denote by C_{00} the number of rows with 0's in both columns, C_{01} the second, C_{10} the third and C_{11} the fourth. Then,

$$(19.2) \quad J(S_{j_1}, S_{j_2}) = \frac{C_{11}}{C_{01} + C_{10} + C_{11}}.$$

To complete the proof by showing that the right-hand side of Equation (19.2) equals $P(x_{j_1}^\pi = x_{j_2}^\pi)$, consider scanning columns j_1, j_2 in increasing row index until the first non-zero entry is found in either column. Because Π is a random permutation, the probability that this smallest row has a 1 in both columns is exactly the right-hand side of Equation (19.2). \square

Thus, our test for the Jaccard coefficient of the shingle sets is probabilistic: we compare the computed values x_i^π from different documents. If a pair coincides, we have candidate near duplicates. Repeat the process independently for 200 random permutations π (a choice suggested in the literature). Call the set of the 200 resulting values of x_i^π the *sketch* $\psi(d_i)$ of d_i . We can then estimate the Jaccard coefficient for any pair of documents d_i, d_j to be $|\psi_i \cap \psi_j|/200$; if this exceeds a preset threshold, we declare that d_i and d_j are similar.

How can we quickly compute $|\psi_i \cap \psi_j|/200$ for all pairs i, j ? Indeed, how do we represent all pairs of documents that are similar, without incurring a blowup that is quadratic in the number of documents? First, we use fingerprints to remove all but one copy of *identical* documents. We may also remove common HTML tags and integers from the shingle computation, to eliminate shingles that occur very commonly in documents without telling us anything about duplication. Next we use a *union-find* algorithm to create clusters that contain documents that are similar. To do this, we must accomplish a crucial step: going from the set of sketches to the set of pairs i, j such that d_i and d_j are similar.

To this end, we compute the number of shingles in common for any pair of documents whose sketches have any members in common. We begin with the list $< x_i^\pi, d_i >$ sorted by x_i^π pairs. For each x_i^π , we can now generate

all pairs i, j for which x_i^π is present in both their sketches. From these we can compute, for each pair i, j with non-zero sketch overlap, a count of the number of x_i^π values they have in common. By applying a preset threshold, we know which pairs i, j have heavily overlapping sketches. For instance, if the threshold were 80%, we would need the count to be at least 160 for any i, j . As we identify such pairs, we run the union-find to group documents into near-duplicate “syntactic clusters”. This is essentially a variant of the single-link clustering algorithm introduced in Section 17.2 (page 382).

One final trick cuts down the space needed in the computation of $|\psi_i \cap \psi_j|/200$ for pairs i, j , which in principle could still demand space quadratic in the number of documents. To remove from consideration those pairs i, j whose sketches have few shingles in common, we preprocess the sketch for each document as follows: sort the x_i^π in the sketch, then shingle this sorted sequence to generate a set of *super-shingles* for each document. If two documents have a super-shingle in common, we proceed to compute the precise value of $|\psi_i \cap \psi_j|/200$. This again is a heuristic but can be highly effective in cutting down the number of i, j pairs for which we accumulate the sketch overlap counts.



Exercise 19.8

Web search engines A and B each crawl a random subset of the same size of the Web. Some of the pages crawled are duplicates – exact textual copies of each other at different URLs. Assume that duplicates are distributed uniformly amongst the pages crawled by A and B. Further, assume that a duplicate is a page that has exactly two copies – no pages have more than two copies. A indexes pages without duplicate elimination whereas B indexes only one copy of each duplicate page. The two random subsets have the same size before duplicate elimination. If, 45% of A’s indexed URLs are present in B’s index, while 50% of B’s indexed URLs are present in A’s index, what fraction of the Web consists of pages that do not have a duplicate?

Exercise 19.9

Instead of using the process depicted in Figure 19.8, consider instead the following process for estimating the Jaccard coefficient of the overlap between two sets S_1 and S_2 . We pick a random subset of the elements of the universe from which S_1 and S_2 are drawn; this corresponds to picking a random subset of the rows of the matrix A in the proof. We exhaustively compute the Jaccard coefficient of these random subsets. Why is this estimate an unbiased estimator of the Jaccard coefficient for S_1 and S_2 ?

Exercise 19.10

Explain why this estimator would be very difficult to use in practice.

19.7 References and further reading

Bush (1945) foreshadowed the Web when he described an information management system that he called *memex*. Berners-Lee et al. (1992) describes one of the earliest incarnations of the Web. Kumar et al. (2000) and Broder

et al. (2000) provide comprehensive studies of the Web as a graph. The use of anchor text was first described in McBryan (1994). The taxonomy of web queries in Section 19.4 is due to Broder (2002). The observation of the power law with exponent 2.1 in Section 19.2.1 appeared in Kumar et al. (1999). Chakrabarti (2002) is a good reference for many aspects of web search and analysis.

The estimation of web search index sizes has a long history of development covered by Bharat and Broder (1998), Lawrence and Giles (1998), Russinovich et al. (2001), Lawrence and Giles (1999), Henzinger et al. (2000), Bar-Yossef and Gurevich (2006). The state of the art is Bar-Yossef and Gurevich (2006), including several of the bias-removal techniques mentioned at the end of Section 19.5. Shingling was introduced by Broder et al. (1997) and used for detecting websites (rather than simply pages) that are identical by Bharat et al. (2000).

20 *Web crawling and indexes*

WEB CRAWLER
SPIDER

20.1 Overview

Web crawling is the process by which we gather pages from the Web, in order to index them and support a search engine. The objective of crawling is to quickly and efficiently gather as many useful web pages as possible, together with the link structure that interconnects them. In Chapter 19 we studied the complexities of the Web stemming from its creation by millions of uncoordinated individuals. In this chapter we study the resulting difficulties for crawling the Web. The focus of this chapter is the component shown in Figure 19.7 as *web crawler*; it is sometimes referred to as a *spider*.

The goal of this chapter is not to describe how to build the crawler for a full-scale commercial web search engine. We focus instead on a range of issues that are generic to crawling from the student project scale to substantial research projects. We begin (Section 20.1.1) by listing desiderata for web crawlers, and then discuss in Section 20.2 how each of these issues is addressed. The remainder of this chapter describes the architecture and some implementation details for a distributed web crawler that satisfies these features. Section 20.3 discusses distributing indexes across many machines for a web-scale implementation.

20.1.1 Features a crawler *must* provide

We list the desiderata for web crawlers in two categories: features that web crawlers *must* provide, followed by features they *should* provide.

Robustness: The Web contains servers that create *spider traps*, which are generators of web pages that mislead crawlers into getting stuck fetching an infinite number of pages in a particular domain. Crawlers must be designed to be resilient to such traps. Not all such traps are malicious; some are the inadvertent side-effect of faulty website development.

Politeness: Web servers have both implicit and explicit policies regulating the rate at which a crawler can visit them. These politeness policies must be respected.

20.1.2 Features a crawler *should* provide

Distributed: The crawler should have the ability to execute in a distributed fashion across multiple machines.

Scalable: The crawler architecture should permit scaling up the crawl rate by adding extra machines and bandwidth.

Performance and efficiency: The crawl system should make efficient use of various system resources including processor, storage and network bandwidth.

Quality: Given that a significant fraction of all web pages are of poor utility for serving user query needs, the crawler should be biased towards fetching “useful” pages first.

Freshness: In many applications, the crawler should operate in continuous mode: it should obtain fresh copies of previously fetched pages. A search engine crawler, for instance, can thus ensure that the search engine’s index contains a fairly current representation of each indexed web page. For such continuous crawling, a crawler should be able to crawl a page with a frequency that approximates the rate of change of that page.

Extensible: Crawlers should be designed to be extensible in many ways – to cope with new data formats, new fetch protocols, and so on. This demands that the crawler architecture be modular.

20.2 Crawling

The basic operation of any hypertext crawler (whether for the Web, an intranet or other hypertext document collection) is as follows. The crawler begins with one or more URLs that constitute a *seed set*. It picks a URL from this seed set, then fetches the web page at that URL. The fetched page is then parsed, to extract both the text and the links from the page (each of which points to another URL). The extracted text is fed to a text indexer (described in Chapters 4 and 5). The extracted links (URLs) are then added to a *URL frontier*, which at all times consists of URLs whose corresponding pages have yet to be fetched by the crawler. Initially, the URL frontier contains the seed set; as pages are fetched, the corresponding URLs are deleted from the URL frontier. The entire process may be viewed as traversing the web graph (see

Chapter 19). In continuous crawling, the URL of a fetched page is added back to the frontier for fetching again in the future.

This seemingly simple recursive traversal of the web graph is complicated by the many demands on a practical web crawling system: the crawler has to be distributed, scalable, efficient, polite, robust and extensible while fetching pages of high quality. We examine the effects of each of these issues. Our treatment follows the design of the *Mercator* crawler that has formed the basis of a number of research and commercial crawlers. As a reference point, fetching a billion pages (a small fraction of the static Web at present) in a month-long crawl requires fetching several hundred pages each second. We will see how to use a multi-threaded design to address several bottlenecks in the overall crawler system in order to attain this fetch rate.

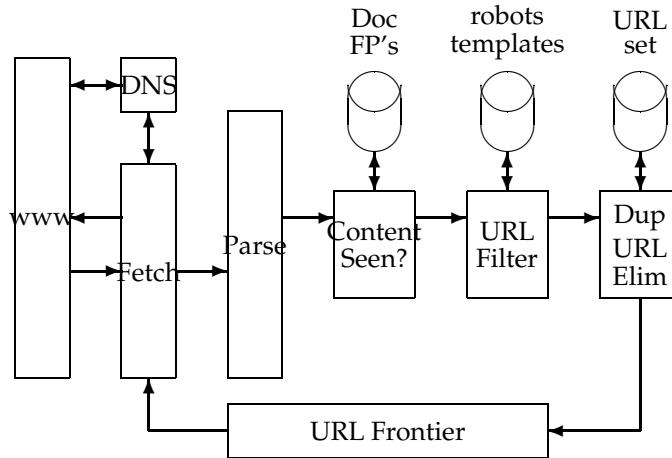
Before proceeding to this detailed description, we reiterate for readers who may attempt to build crawlers of some basic properties any non-professional crawler should satisfy:

1. Only one connection should be open to any given host at a time.
2. A waiting time of a few seconds should occur between successive requests to a host.
3. Politeness restrictions detailed in Section 20.2.1 should be obeyed.

20.2.1 Crawler architecture

The simple scheme outlined above for crawling demands several modules that fit together as shown in Figure 20.1.

1. The URL frontier, containing URLs yet to be fetched in the current crawl (in the case of continuous crawling, a URL may have been fetched previously but is back in the frontier for re-fetching). We describe this further in Section 20.2.3.
2. A *DNS resolution* module that determines the web server from which to fetch the page specified by a URL. We describe this further in Section 20.2.2.
3. A fetch module that uses the http protocol to retrieve the web page at a URL.
4. A parsing module that extracts the text and set of links from a fetched web page.
5. A duplicate elimination module that determines whether an extracted link is already in the URL frontier or has recently been fetched.



► **Figure 20.1** The basic crawler architecture.

Crawling is performed by anywhere from one to potentially hundreds of threads, each of which loops through the logical cycle in Figure 20.1. These threads may be run in a single process, or be partitioned amongst multiple processes running at different nodes of a distributed system. We begin by assuming that the URL frontier is in place and non-empty and defer our description of the implementation of the URL frontier to Section 20.2.3. We follow the progress of a single URL through the cycle of being fetched, passing through various checks and filters, then finally (for continuous crawling) being returned to the URL frontier.

A crawler thread begins by taking a URL from the frontier and fetching the web page at that URL, generally using the http protocol. The fetched page is then written into a temporary store, where a number of operations are performed on it. Next, the page is parsed and the text as well as the links in it are extracted. The text (with any tag information – e.g., terms in boldface) is passed on to the indexer. Link information including anchor text is also passed on to the indexer for use in ranking in ways that are described in Chapter 21. In addition, each extracted link goes through a series of tests to determine whether the link should be added to the URL frontier.

First, the thread tests whether a web page with the same content has already been seen at another URL. The simplest implementation for this would use a simple fingerprint such as a checksum (placed in a store labeled "Doc FP's" in Figure 20.1). A more sophisticated test would use shingles instead

of fingerprints, as described in Chapter 19.

Next, a *URL filter* is used to determine whether the extracted URL should be excluded from the frontier based on one of several tests. For instance, the crawl may seek to exclude certain domains (say, all .com URLs) – in this case the test would simply filter out the URL if it were from the .com domain. A similar test could be inclusive rather than exclusive. Many hosts on the Web place certain portions of their websites off-limits to crawling, under a standard known as the *Robots Exclusion Protocol*. This is done by placing a file with the name robots.txt at the root of the URL hierarchy at the site. Here is an example robots.txt file that specifies that no robot should visit any URL whose position in the file hierarchy starts with /yoursite/temp/, except for the robot called “searchengine”.

```
User-agent: *
Disallow: /yoursite/temp/

User-agent: searchengine
Disallow:
```

The robots.txt file must be fetched from a website in order to test whether the URL under consideration passes the robot restrictions, and can therefore be added to the URL frontier. Rather than fetch it afresh for testing on each URL to be added to the frontier, a cache can be used to obtain a recently fetched copy of the file for the host. This is especially important since many of the links extracted from a page fall within the host from which the page was fetched and therefore can be tested against the host’s robots.txt file. Thus, by performing the filtering during the link extraction process, we would have especially high locality in the stream of hosts that we need to test for robots.txt files, leading to high cache hit rates. Unfortunately, this runs afoul of webmasters’ politeness expectations. A URL (particularly one referring to a low-quality or rarely changing document) may be in the frontier for days or even weeks. If we were to perform the robots filtering *before* adding such a URL to the frontier, its robots.txt file could have changed by the time the URL is dequeued from the frontier and fetched. We must consequently perform robots-filtering immediately before attempting to fetch a web page. As it turns out, maintaining a cache of robots.txt files is still highly effective; there is sufficient locality even in the stream of URLs dequeued from the URL frontier.

Next, a URL should be *normalized* in the following sense: often the HTML encoding of a link from a web page p indicates the target of that link relative to the page p . Thus, there is a relative link encoded thus in the HTML of the page en.wikipedia.org/wiki/Main_Page:

```
<a href="/wiki/Wikipedia:General_disclaimer" title="Wikipedia:General
disclaimer">Disclaimers</a>
```

points to the URL http://en.wikipedia.org/wiki/Wikipedia:General_disclaimer.

Finally, the URL is checked for duplicate elimination: if the URL is already in the frontier or (in the case of a non-continuous crawl) already crawled, we do not add it to the frontier. When the URL is added to the frontier, it is assigned a priority based on which it is eventually removed from the frontier for fetching. The details of this priority queuing are in Section 20.2.3.

Certain housekeeping tasks are typically performed by a dedicated thread. This thread is generally quiescent except that it wakes up once every few seconds to log crawl progress statistics (URLs crawled, frontier size, etc.), decide whether to terminate the crawl, or (once every few hours of crawling) checkpoint the crawl. In checkpointing, a snapshot of the crawler's state (say, the URL frontier) is committed to disk. In the event of a catastrophic crawler failure, the crawl is restarted from the most recent checkpoint.

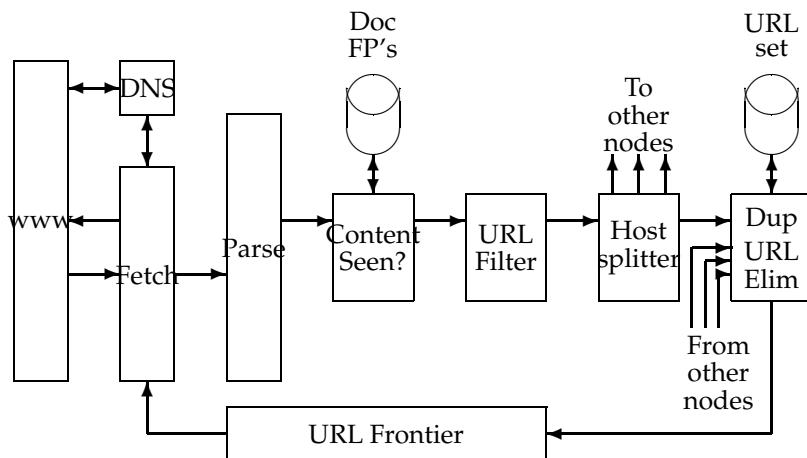
Distributing the crawler

We have mentioned that the threads in a crawler could run under different processes, each at a different node of a distributed crawling system. Such distribution is essential for scaling; it can also be of use in a geographically distributed crawler system where each node crawls hosts "near" it. Partitioning the hosts being crawled amongst the crawler nodes can be done by a hash function, or by some more specifically tailored policy. For instance, we may locate a crawler node in Europe to focus on European domains, although this is not dependable for several reasons – the routes that packets take through the internet do not always reflect geographic proximity, and in any case the domain of a host does not always reflect its physical location.

How do the various nodes of a distributed crawler communicate and share URLs? The idea is to replicate the flow of Figure 20.1 at each node, with one essential difference: following the URL filter, we use a *host splitter* to dispatch each surviving URL to the crawler node responsible for the URL; thus the set of hosts being crawled is partitioned among the nodes. This modified flow is shown in Figure 20.2. The output of the host splitter goes into the Duplicate URL Eliminator block of each other node in the distributed system.

The "Content Seen?" module in the distributed architecture of Figure 20.2 is, however, complicated by several factors:

1. Unlike the URL frontier and the duplicate elimination module, document fingerprints/shingles cannot be partitioned based on host name. There is nothing preventing the same (or highly similar) content from appearing on different web servers. Consequently, the set of fingerprints/shingles must be partitioned across the nodes based on some property of the fin-



► Figure 20.2 Distributing the basic crawl architecture.

gerprint/shingle (say by taking the fingerprint modulo the number of nodes). The result of this locality-mismatch is that most “Content Seen?” tests result in a remote procedure call (although it is possible to batch lookup requests).

2. There is very little locality in the stream of document fingerprints/shingles. Thus, caching popular fingerprints does not help (since there are no popular fingerprints).
3. Documents change over time and so, in the context of continuous crawling, we must be able to delete their outdated fingerprints/shingles from the content-seen set(s). In order to do so, it is necessary to save the finger-print/shingle of the document in the URL frontier, along with the URL itself.

20.2.2 DNS resolution

IP ADDRESS Each web server (and indeed any host connected to the internet) has a unique *IP address*: a sequence of four bytes generally represented as four integers separated by dots; for instance 207.142.131.248 is the numerical IP address associated with the host www.wikipedia.org. Given a URL such as www.wikipedia.org in textual form, translating it to an IP address (in this case, 207.142.131.248) is

DNS RESOLUTION

a process known as *DNS resolution* or DNS lookup; here DNS stands for *Domain Name Service*. During DNS resolution, the program that wishes to perform this translation (in our case, a component of the web crawler) contacts a *DNS server* that returns the translated IP address. (In practice the entire translation may not occur at a single DNS server; rather, the DNS server contacted initially may recursively call upon other DNS servers to complete the translation.) For a more complex URL such as en.wikipedia.org/wiki/Domain_Name_System, the crawler component responsible for DNS resolution extracts the host name – in this case en.wikipedia.org – and looks up the IP address for the host en.wikipedia.org.

DNS resolution is a well-known bottleneck in web crawling. Due to the distributed nature of the Domain Name Service, DNS resolution may entail multiple requests and round-trips across the internet, requiring seconds and sometimes even longer. Right away, this puts in jeopardy our goal of fetching several hundred documents a second. A standard remedy is to introduce caching: URLs for which we have recently performed DNS lookups are likely to be found in the DNS cache, avoiding the need to go to the DNS servers on the internet. However, obeying politeness constraints (see Section 20.2.3) limits the of cache hit rate.

There is another important difficulty in DNS resolution; the lookup implementations in standard libraries (likely to be used by anyone developing a crawler) are generally synchronous. This means that once a request is made to the Domain Name Service, other crawler threads at that node are blocked until the first request is completed. To circumvent this, most web crawlers implement their own DNS resolver as a component of the crawler. Thread i executing the resolver code sends a message to the DNS server and then performs a timed wait: it resumes either when being signaled by another thread or when a set time quantum expires. A single, separate DNS thread listens on the standard DNS port (port 53) for incoming response packets from the name service. Upon receiving a response, it signals the appropriate crawler thread (in this case, i) and hands it the response packet if i has not yet resumed because its time quantum has expired. A crawler thread that resumes because its wait time quantum has expired retries for a fixed number of attempts, sending out a new message to the DNS server and performing a timed wait each time; the designers of Mercator recommend of the order of five attempts. The time quantum of the wait increases exponentially with each of these attempts; Mercator started with one second and ended with roughly 90 seconds, in consideration of the fact that there are host names that take tens of seconds to resolve.

20.2.3 The URL frontier

The URL frontier at a node is given a URL by its crawl process (or by the host splitter of another crawl process). It maintains the URLs in the frontier and regurgitates them in some order whenever a crawler thread seeks a URL. Two important considerations govern the order in which URLs are returned by the frontier. First, high-quality pages that change frequently should be prioritized for frequent crawling. Thus, the priority of a page should be a function of both its change rate and its quality (using some reasonable quality estimate). The combination is necessary because a large number of spam pages change completely on every fetch.

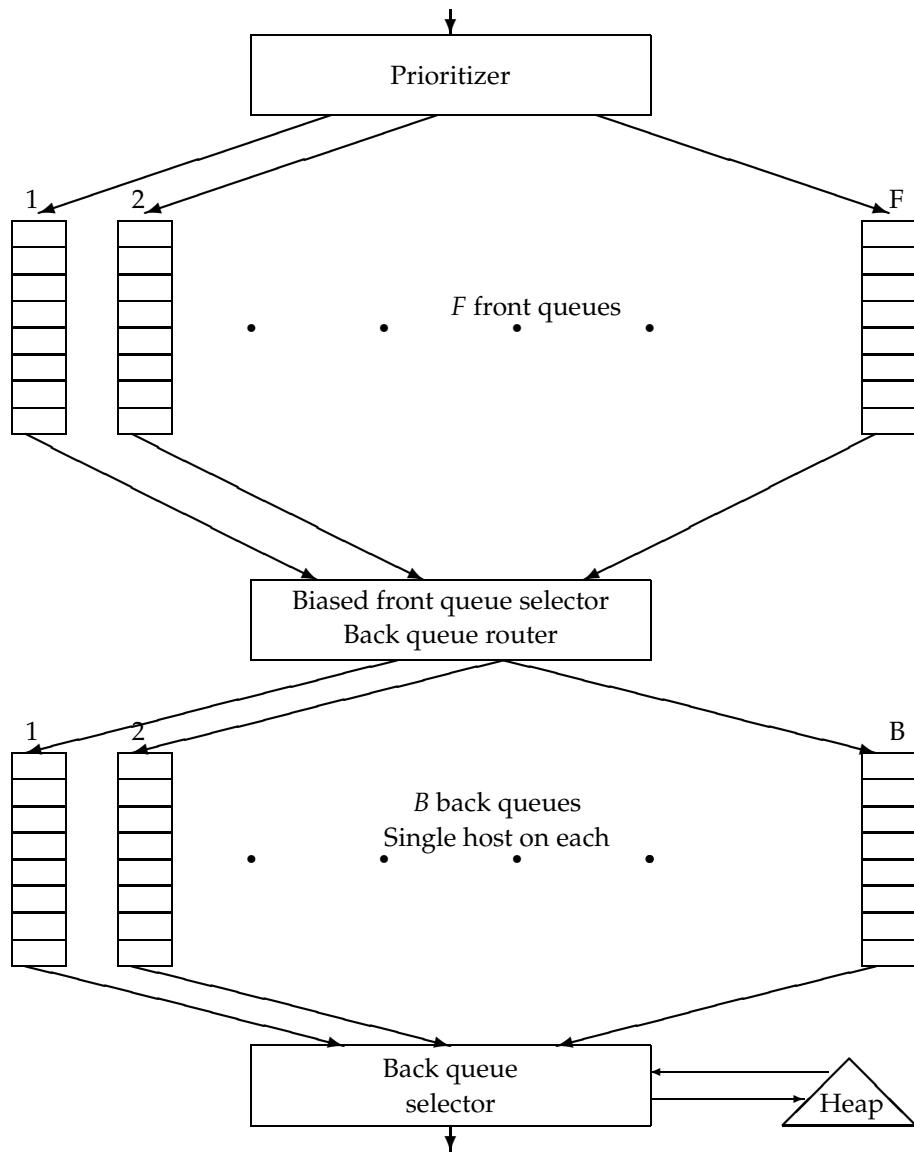
The second consideration is politeness: we must avoid repeated fetch requests to a host within a short time span. The likelihood of this is exacerbated because of a form of locality of reference: many URLs link to other URLs at the same host. As a result, a URL frontier implemented as a simple priority queue might result in a burst of fetch requests to a host. This might occur even if we were to constrain the crawler so that at most one thread could fetch from any single host at any time. A common heuristic is to insert a gap between successive fetch requests to a host that is an order of magnitude larger than the time taken for the most recent fetch from that host.

Figure 20.3 shows a polite and prioritizing implementation of a URL frontier. Its goals are to ensure that (i) only one connection is open at a time to any host; (ii) a waiting time of a few seconds occurs between successive requests to a host and (iii) high-priority pages are crawled preferentially.

The two major sub-modules are a set of *F front queues* in the upper portion of the figure, and a set of *B back queues* in the lower part; all of these are FIFO queues. The front queues implement the prioritization, while the back queues implement politeness. In the flow of a URL added to the frontier as it makes its way through the front and back queues, a *prioritizer* first assigns to the URL an integer priority i between 1 and F based on its fetch history (taking into account the rate at which the web page at this URL has changed between previous crawls). For instance, a document that has exhibited frequent change would be assigned a higher priority. Other heuristics could be application-dependent and explicit – for instance, URLs from news services may always be assigned the highest priority. Now that it has been assigned priority i , the URL is now appended to the i th of the front queues.

Each of the B back queues maintains the following invariants: (i) it is non-empty while the crawl is in progress and (ii) it only contains URLs from a single host¹. An auxiliary table T (Figure 20.4) is used to maintain the mapping from hosts to back queues. Whenever a back-queue is empty and is being re-filled from a front-queue, table T must be updated accordingly.

1. The number of hosts is assumed to far exceed B .



► **Figure 20.3** The URL frontier. URLs extracted from already crawled pages flow in at the top of the figure. A crawl thread requesting a URL extracts it from the bottom of the figure. En route, a URL flows through one of several *front queues* that manage its priority for crawling, followed by one of several *back queues* that manage the crawler's politeness.

Host	Back queue
stanford.edu	23
microsoft.com	47
acm.org	12

► **Figure 20.4** Example of an auxiliary hosts-to-back queues table.

In addition, we maintain a heap with one entry for each back queue, the entry being the earliest time t_e at which the host corresponding to that queue can be contacted again.

A crawler thread requesting a URL from the frontier extracts the root of this heap and (if necessary) waits until the corresponding time entry t_e . It then takes the URL u at the head of the back queue j corresponding to the extracted heap root, and proceeds to fetch the URL u . After fetching u , the calling thread checks whether j is empty. If so, it picks a front queue and extracts from its head a URL v . The choice of front queue is biased (usually by a random process) towards queues of higher priority, ensuring that URLs of high priority flow more quickly into the back queues. We examine v to check whether there is already a back queue holding URLs from its host. If so, v is added to that queue and we reach back to the front queues to find another candidate URL for insertion into the now-empty queue j . This process continues until j is non-empty again. In any case, the thread inserts a heap entry for j with a new earliest time t_e based on the properties of the URL in j that was last fetched (such as when its host was last contacted as well as the time taken for the last fetch), then continues with its processing. For instance, the new entry t_e could be the current time plus ten times the last fetch time.

The number of front queues, together with the policy of assigning priorities and picking queues, determines the priority properties we wish to build into the system. The number of back queues governs the extent to which we can keep all crawl threads busy while respecting politeness. The designers of Mercator recommend a rough rule of three times as many back queues as crawler threads.

On a Web-scale crawl, the URL frontier may grow to the point where it demands more memory at a node than is available. The solution is to let most of the URL frontier reside on disk. A portion of each queue is kept in memory, with more brought in from disk as it is drained in memory.



Exercise 20.1

Why is it better to partition hosts (rather than individual URLs) between the nodes of a distributed crawl system?

Exercise 20.2

Why should the host splitter precede the Duplicate URL Eliminator?

Exercise 20.3

[***]

In the preceding discussion we encountered two recommended “hard constants” – the increment on t_e being ten times the last fetch time, and the number of back queues being three times the number of crawl threads. How are these two constants related?

20.3 Distributing indexes

In Section 4.4 we described distributed indexing. We now consider the distribution of the index across a large computer cluster² that supports querying. Two obvious alternative index implementations suggest themselves: *partitioning by terms*, also known as global index organization, and *partitioning by documents*, also known as local index organization. In the former, the dictionary of index terms is partitioned into subsets, each subset residing at a node. Along with the terms at a node, we keep the postings for those terms. A query is routed to the nodes corresponding to its query terms. In principle, this allows greater concurrency since a stream of queries with different query terms would hit different sets of machines.

In practice, partitioning indexes by vocabulary terms turns out to be non-trivial. Multi-word queries require the sending of long postings lists between sets of nodes for merging, and the cost of this can outweigh the greater concurrency. Load balancing the partition is governed not by an a priori analysis of relative term frequencies, but rather by the distribution of query terms and their co-occurrences, which can drift with time or exhibit sudden bursts. Achieving good partitions is a function of the co-occurrences of query terms and entails the clustering of terms to optimize objectives that are not easy to quantify. Finally, this strategy makes implementation of dynamic indexing more difficult.

A more common implementation is to partition by documents: each node contains the index for a subset of all documents. Each query is distributed to all nodes, with the results from various nodes being merged before presentation to the user. This strategy trades more local disk seeks for less inter-node communication. One difficulty in this approach is that global statistics used in scoring – such as idf – must be computed across the entire document collection even though the index at any single node only contains a subset of the documents. These are computed by distributed “background” processes that periodically refresh the node indexes with fresh global statistics.

How do we decide the partition of documents to nodes? Based on our development of the crawler architecture in Section 20.2.1, one simple approach would be to assign all pages from a host to a single node. This partitioning

2. Please note the different usage of “clusters” elsewhere in this book, in the sense of Chapters 16 and 17.

could follow the partitioning of hosts to crawler nodes. A danger of such partitioning is that on many queries, a preponderance of the results would come from documents at a small number of hosts (and hence a small number of index nodes).

A hash of each URL into the space of index nodes results in a more uniform distribution of query-time computation across nodes. At query time, the query is broadcast to each of the nodes, with the top k results from each node being merged to find the top k documents for the query. A common implementation heuristic is to partition the document collection into indexes of documents that are more likely to score highly on most queries (using, for instance, techniques in Chapter 21) and low-scoring indexes with the remaining documents. We only search the low-scoring indexes when there are too few matches in the high-scoring indexes, as described in Section 7.2.1.

20.4 Connectivity servers

CONNECTIVITY SERVER
CONNECTIVITY
QUERIES

For reasons to become clearer in Chapter 21, web search engines require a *connectivity server* that supports fast *connectivity queries* on the web graph. Typical connectivity queries are *which URLs link to a given URL?* and *which URLs does a given URL link to?* To this end, we wish to store mappings in memory from URL to out-links, and from URL to in-links. Applications include crawl control, web graph analysis, sophisticated crawl optimization and *link analysis* (to be covered in Chapter 21).

Suppose that the Web had four billion pages, each with ten links to other pages. In the simplest form, we would require 32 bits or 4 bytes to specify each end (source and destination) of each link, requiring a total of

$$4 \times 10^9 \times 10 \times 8 = 3.2 \times 10^{11}$$

bytes of memory. Some basic properties of the web graph can be exploited to use well under 10% of this memory requirement. At first sight, we appear to have a data compression problem – which is amenable to a variety of standard solutions. However, our goal is not to simply compress the web graph to fit into memory; we must do so in a way that efficiently supports connectivity queries; this challenge is reminiscent of index compression (Chapter 5).

We assume that each web page is represented by a unique integer; the specific scheme used to assign these integers is described below. We build an *adjacency table* that resembles an inverted index: it has a row for each web page, with the rows ordered by the corresponding integers. The row for any page p contains a sorted list of integers, each corresponding to a web page that links to p . This table permits us to respond to queries of the form *which pages link to p ?* In similar fashion we build a table whose entries are the pages linked to by p .

```

1: www.stanford.edu/alchemy
2: www.stanford.edu/biology
3: www.stanford.edu/biology/plant
4: www.stanford.edu/biology/plant/copyright
5: www.stanford.edu/biology/plant/people
6: www.stanford.edu/chemistry

```

► **Figure 20.5** A lexicographically ordered set of URLs.

This table representation cuts the space taken by the naive representation (in which we explicitly represent each link by its two end points, each a 32-bit integer) by 50%. Our description below will focus on the table for the links *from* each page; it should be clear that the techniques apply just as well to the table of links to each page. To further reduce the storage for the table, we exploit several ideas:

1. Similarity between lists: Many rows of the table have many entries in common. Thus, if we explicitly represent a prototype row for several similar rows, the remainder can be succinctly expressed in terms of the prototypical row.
2. Locality: many links from a page go to “nearby” pages – pages on the same host, for instance. This suggests that in encoding the destination of a link, we can often use small integers and thereby save space.
3. We use gap encodings in sorted lists: rather than store the destination of each link, we store the offset from the previous entry in the row.

We now develop each of these techniques.

In a *lexicographic* ordering of all URLs, we treat each URL as an alphanumeric string and sort these strings. Figure 20.5 shows a segment of this sorted order. For a true lexicographic sort of web pages, the domain name part of the URL should be inverted, so that `www.stanford.edu` becomes `edu.stanford.www`, but this is not necessary here since we are mainly concerned with links local to a single host.

To each URL, we assign its position in this ordering as the unique identifying integer. Figure 20.6 shows an example of such a numbering and the resulting table. In this example sequence, `www.stanford.edu/biology` is assigned the integer 2 since it is second in the sequence.

We next exploit a property that stems from the way most websites are structured to get similarity and locality. Most websites have a template with a set of links from each page in the site to a fixed set of pages on the site (such

```

1: 1, 2, 4, 8, 16, 32, 64
2: 1, 4, 9, 16, 25, 36, 49, 64
3: 1, 2, 3, 5, 8, 13, 21, 34, 55, 89, 144
4: 1, 4, 8, 16, 25, 36, 49, 64

```

► **Figure 20.6** A four-row segment of the table of links.

as its copyright notice, terms of use, and so on). In this case, the rows corresponding to pages in a website will have many table entries in common. Moreover, under the lexicographic ordering of URLs, it is very likely that the pages from a website appear as contiguous rows in the table.

We adopt the following strategy: we walk down the table, encoding each table row in terms of the seven preceding rows. In the example of Figure 20.6, we could encode the fourth row as “the same as the row at offset 2 (meaning, two rows earlier in the table), with 9 replaced by 8”. This requires the specification of the offset, the integer(s) dropped (in this case 9) and the integer(s) added (in this case 8). The use of only the seven preceding rows has two advantages: (i) the offset can be expressed with only 3 bits; this choice is optimized empirically (the reason for seven and not eight preceding rows is the subject of Exercise 20.4) and (ii) fixing the maximum offset to a small value like seven avoids having to perform an expensive search among many candidate prototypes in terms of which to express the current row.

What if none of the preceding seven rows is a good prototype for expressing the current row? This would happen, for instance, at each boundary between different websites as we walk down the rows of the table. In this case we simply express the row as starting from the empty set and “adding in” each integer in that row. By using gap encodings to store the gaps (rather than the actual integers) in each row, and encoding these gaps tightly based on the distribution of their values, we obtain further space reduction. In experiments mentioned in Section 20.5, the series of techniques outlined here appears to use as few as 3 bits per link, on average – a dramatic reduction from the 64 required in the naive representation.

While these ideas give us a representation of sizable web graphs that comfortably fit in memory, we still need to support connectivity queries. What is entailed in retrieving from this representation the set of links from a page? First, we need an index lookup from (a hash of) the URL to its row number in the table. Next, we need to reconstruct these entries, which may be encoded in terms of entries in other rows. This entails following the offsets to reconstruct these other rows – a process that in principle could lead through many levels of indirection. In practice however, this does not happen very often. A heuristic for controlling this can be introduced into the construc-

tion of the table: when examining the preceding seven rows as candidates from which to model the current row, we demand a threshold of similarity between the current row and the candidate prototype. This threshold must be chosen with care. If the threshold is set too high, we seldom use prototypes and express many rows afresh. If the threshold is too low, most rows get expressed in terms of prototypes, so that at query time the reconstruction of a row leads to many levels of indirection through preceding prototypes.



Exercise 20.4

We noted that expressing a row in terms of one of seven preceding rows allowed us to use no more than three bits to specify which of the preceding rows we are using as prototype. Why seven and not eight preceding rows? (*Hint: consider the case when none of the preceding seven rows is a good prototype.*)

Exercise 20.5

We noted that for the scheme in Section 20.4, decoding the links incident on a URL could result in many levels of indirection. Construct an example in which the number of levels of indirection grows linearly with the number of URLs.

20.5 References and further reading

The first web crawler appears to be Matthew Gray's Wanderer, written in the spring of 1993. The Mercator crawler is due to Najork and Heydon (Najork and Heydon 2001; 2002); the treatment in this chapter follows their work. Other classic early descriptions of web crawling include Burner (1997), Brin and Page (1998), Cho et al. (1998) and the creators of the Webbase system at Stanford (Hirai et al. 2000). Cho and Garcia-Molina (2002) give a taxonomy and comparative study of different modes of communication between the nodes of a distributed crawler. The Robots Exclusion Protocol standard is described at <http://www.robotstxt.org/wc/exclusion.html>. Boldi et al. (2002) and Shkapenyuk and Suel (2002) provide more recent details of implementing large-scale distributed web crawlers.

Our discussion of DNS resolution (Section 20.2.2) uses the current convention for internet addresses, known as IPv4 (for Internet Protocol version 4) – each IP address is a sequence of four bytes. In the future, the convention for addresses (collectively known as the internet *address space*) is likely to use a new standard known as IPv6 (<http://www.ipv6.org/>).

Tomasic and Garcia-Molina (1993) and Jeong and Omiecinski (1995) are key early papers evaluating term partitioning versus document partitioning for distributed indexes. Document partitioning is found to be superior, at least when the distribution of terms is skewed, as it typically is in practice. This result has generally been confirmed in more recent work (MacFarlane et al. 2000). But the outcome depends on the details of the distributed system;

at least one thread of work has reached the opposite conclusion (Ribeiro-Neto and Barbosa 1998, Badue et al. 2001). Sornil (2001) argues for a partitioning scheme that is a hybrid between term and document partitioning. Barroso et al. (2003) describe the distribution methods used at Google. The first implementation of a connectivity server was described by Bharat et al. (1998). The scheme discussed in this chapter, currently believed to be the best published scheme (achieving as few as 3 bits per link for encoding), is described in a series of papers by Boldi and Vigna (2004a;b).

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21

Link analysis

The analysis of hyperlinks and the graph structure of the Web has been instrumental in the development of web search. In this chapter we focus on the use of hyperlinks for ranking web search results. Such link analysis is one of many factors considered by web search engines in computing a composite score for a web page on any given query. We begin by reviewing some basics of the Web as a graph in Section 21.1, then proceed to the technical development of the elements of link analysis for ranking.

Link analysis for web search has intellectual antecedents in the field of citation analysis, aspects of which overlap with an area known as bibliometrics. These disciplines seek to quantify the influence of scholarly articles by analyzing the pattern of citations amongst them. Much as citations represent the conferral of authority from a scholarly article to others, link analysis on the Web treats hyperlinks from a web page to another as a conferral of authority. Clearly, not every citation or hyperlink implies such authority conferral; for this reason, simply measuring the quality of a web page by the number of in-links (citations from other pages) is not robust enough. For instance, one may contrive to set up multiple web pages pointing to a target web page, with the intent of artificially boosting the latter's tally of in-links. This phenomenon is referred to as link spam. Nevertheless, the phenomenon of citation is prevalent and dependable enough that it is feasible for web search engines to derive useful signals for ranking from more sophisticated link analysis. Link analysis also proves to be a useful indicator of what page(s) to crawl next while crawling the web; this is done by using link analysis to guide the priority assignment in the front queues of Chapter 20.

Section 21.1 develops the basic ideas underlying the use of the web graph in link analysis. Sections 21.2 and 21.3 then develop two distinct methods for link analysis, PageRank and HITS.

21.1 The Web as a graph

Recall the notion of the web graph from Section 19.2.1 and particularly Figure 19.2. Our study of link analysis builds on two intuitions:

1. The anchor text pointing to page B is a good description of page B.
2. The hyperlink from A to B represents an endorsement of page B, by the creator of page A. This is not always the case; for instance, many links amongst pages within a single website stem from the user of a common template. For instance, most corporate websites have a pointer from every page to a page containing a copyright notice – this is clearly not an endorsement. Accordingly, implementations of link analysis algorithms will typically discount such “internal” links.

21.1.1 Anchor text and the web graph

The following fragment of HTML code from a web page shows a hyperlink pointing to the home page of the Journal of the ACM:

```
<a href="http://www.acm.org/jacm/">Journal of the ACM.</a>
```

In this case, the link points to the page `http://www.acm.org/jacm/` and the anchor text is *Journal of the ACM*. Clearly, in this example the anchor is descriptive of the target page. But then the target page (B = `http://www.acm.org/jacm/`) itself contains the same description as well as considerable additional information on the journal. So what use is the anchor text?

The Web is full of instances where the page B does not provide an accurate description of itself. In many cases this is a matter of how the publishers of page B choose to present themselves; this is especially common with corporate web pages, where a web presence is a marketing statement. For example, at the time of the writing of this book the home page of the IBM corporation (`http://www.ibm.com`) did not contain the term computer anywhere in its HTML code, despite the fact that IBM is widely viewed as the world’s largest computer maker. Similarly, the HTML code for the home page of Yahoo! (`http://www.yahoo.com`) does not at this time contain the word portal.

Thus, there is often a gap between the terms in a web page, and how web users would describe that web page. Consequently, web searchers need not use the terms in a page to query for it. In addition, many web pages are rich in graphics and images, and/or embed their text in these images; in such cases, the HTML parsing performed when crawling will not extract text that is useful for indexing these pages. The “standard IR” approach to this would be to use the methods outlined in Chapter 9 and Section 12.4. The insight

behind anchor text is that such methods can be supplanted by anchor text, thereby tapping the power of the community of web page authors.

The fact that the anchors of many hyperlinks pointing to `http://www.ibm.com` include the word computer can be exploited by web search engines. For instance, the anchor text terms can be included as terms under which to index the target web page. Thus, the postings for the term computer would include the document `http://www.ibm.com` and that for the term portal would include the document `http://www.yahoo.com`, using a special indicator to show that these terms occur as anchor (rather than in-page) text. As with in-page terms, anchor text terms are generally weighted based on frequency, with a penalty for terms that occur very often (the most common terms in anchor text across the Web are Click and here, using methods very similar to idf). The actual weighting of terms is determined by machine-learned scoring, as in Section 15.4.1; current web search engines appear to assign a substantial weighting to anchor text terms.

The use of anchor text has some interesting side-effects. Searching for big blue on most web search engines returns the home page of the IBM corporation as the top hit; this is consistent with the popular nickname that many people use to refer to IBM. On the other hand, there have been (and continue to be) many instances where derogatory anchor text such as evil empire leads to somewhat unexpected results on querying for these terms on web search engines. This phenomenon has been exploited in orchestrated campaigns against specific sites. Such orchestrated anchor text may be a form of spamming, since a website can create misleading anchor text pointing to itself, to boost its ranking on selected query terms. Detecting and combating such systematic abuse of anchor text is another form of spam detection that web search engines perform.

The window of text surrounding anchor text (sometimes referred to as *extended anchor text*) is often usable in the same manner as anchor text itself; consider for instance the fragment of web text `there is good discussion of vedic scripture <a>here`. This has been considered in a number of settings and the useful width of this window has been studied; see Section 21.4 for references.

**Exercise 21.1**

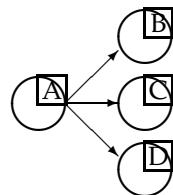
Is it always possible to follow directed edges (hyperlinks) in the web graph from any node (web page) to any other? Why or why not?

Exercise 21.2

Find an instance of misleading anchor-text on the Web.

Exercise 21.3

Given the collection of anchor-text phrases for a web page x , suggest a heuristic for choosing one term or phrase from this collection that is most descriptive of x .



► **Figure 21.1** The random surfer at node A proceeds with probability 1/3 to each of B, C and D.

Exercise 21.4

Does your heuristic in the previous exercise take into account a single domain D repeating anchor text for x from multiple pages in D ?

21.2 PageRank

PAGERANK

We now focus on scoring and ranking measures derived from the link structure alone. Our first technique for link analysis assigns to every node in the web graph a numerical score between 0 and 1, known as its *PageRank*. The PageRank of a node will depend on the link structure of the web graph. Given a query, a web search engine computes a composite score for each web page that combines hundreds of features such as cosine similarity (Section 6.3) and term proximity (Section 7.2.2), together with the PageRank score. This composite score, developed using the methods of Section 15.4.1, is used to provide a ranked list of results for the query.

Consider a random surfer who begins at a web page (a node of the web graph) and executes a random walk on the Web as follows. At each time step, the surfer proceeds from his current page A to a randomly chosen web page that A hyperlinks to. Figure 21.1 shows the surfer at a node A, out of which there are three hyperlinks to nodes B, C and D; the surfer proceeds at the next time step to one of these three nodes, with equal probabilities 1/3.

As the surfer proceeds in this random walk from node to node, he visits some nodes more often than others; intuitively, these are nodes with many links coming in from other frequently visited nodes. The idea behind PageRank is that pages visited more often in this walk are more important.

TELEPORT

What if the current location of the surfer, the node A, has no out-links? To address this we introduce an additional operation for our random surfer: the *teleport* operation. In the teleport operation the surfer jumps from a node to any other node in the web graph. This could happen because he types

an address into the URL bar of his browser. The destination of a teleport operation is modeled as being chosen uniformly at random from all web pages. In other words, if N is the total number of nodes in the web graph¹, the teleport operation takes the surfer to each node with probability $1/N$. The surfer would also teleport to his present position with probability $1/N$.

In assigning a PageRank score to each node of the web graph, we use the teleport operation in two ways: (1) When at a node with no out-links, the surfer invokes the teleport operation. (2) At any node that has outgoing links, the surfer invokes the teleport operation with probability $0 < \alpha < 1$ and the standard random walk (follow an out-link chosen uniformly at random as in Figure 21.1) with probability $1 - \alpha$, where α is a fixed parameter chosen in advance. Typically, α might be 0.1.

In Section 21.2.1, we will use the theory of Markov chains to argue that when the surfer follows this combined process (random walk plus teleport) he visits each node v of the web graph a fixed fraction of the time $\pi(v)$ that depends on (1) the structure of the web graph and (2) the value of α . We call this value $\pi(v)$ the PageRank of v and will show how to compute this value in Section 21.2.2.

21.2.1 Markov chains

A Markov chain is a *discrete-time stochastic process*: a process that occurs in a series of time-steps in each of which a random choice is made. A Markov chain consists of N states. Each web page will correspond to a state in the Markov chain we will formulate.

A Markov chain is characterized by an $N \times N$ transition probability matrix P each of whose entries is in the interval $[0, 1]$; the entries in each row of P add up to 1. The Markov chain can be in one of the N states at any given time-step; then, the entry P_{ij} tells us the probability that the state at the next time-step is j , conditioned on the current state being i . Each entry P_{ij} is known as a transition probability and depends only on the current state i ; this is known as the Markov property. Thus, by the Markov property,

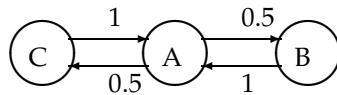
$$\forall i, j, P_{ij} \in [0, 1]$$

and

$$(21.1) \quad \forall i, \sum_{j=1}^N P_{ij} = 1.$$

A matrix with non-negative entries that satisfies Equation (21.1) is known as a *stochastic matrix*. A key property of a stochastic matrix is that it has a *principal left eigenvector* corresponding to its largest eigenvalue, which is 1.

1. This is consistent with our usage of N for the number of documents in the collection.



► **Figure 21.2** A simple Markov chain with three states; the numbers on the links indicate the transition probabilities.

In a Markov chain, the probability distribution of next states for a Markov chain depends only on the current state, and not on how the Markov chain arrived at the current state. Figure 21.2 shows a simple Markov chain with three states. From the middle state A, we proceed with (equal) probabilities of 0.5 to either B or C. From either B or C, we proceed with probability 1 to A. The transition probability matrix of this Markov chain is then

$$\begin{pmatrix} 0 & 0.5 & 0.5 \\ 1 & 0 & 0 \\ 1 & 0 & 0 \end{pmatrix}$$

PROBABILITY VECTOR

A Markov chain's probability distribution over its states may be viewed as a *probability vector*: a vector all of whose entries are in the interval $[0, 1]$, and the entries add up to 1. An N -dimensional probability vector each of whose components corresponds to one of the N states of a Markov chain can be viewed as a probability distribution over its states. For our simple Markov chain of Figure 21.2, the probability vector would have 3 components that sum to 1.

We can view a random surfer on the web graph as a Markov chain, with one state for each web page, and each transition probability representing the probability of moving from one web page to another. The teleport operation contributes to these transition probabilities. The adjacency matrix A of the web graph is defined as follows: if there is a hyperlink from page i to page j , then $A_{ij} = 1$, otherwise $A_{ij} = 0$. We can readily derive the transition probability matrix P for our Markov chain from the $N \times N$ matrix A :

1. If a row of A has no 1's, then replace each element by $1/N$. For all other rows proceed as follows.
2. Divide each 1 in A by the number of 1's in its row. Thus, if there is a row with three 1's, then each of them is replaced by $1/3$.
3. Multiply the resulting matrix by $1 - \alpha$.

4. Add α/N to every entry of the resulting matrix, to obtain P .

We can depict the probability distribution of the surfer's position at any time by a probability vector \vec{x} . At $t = 0$ the surfer may begin at a state whose corresponding entry in \vec{x} is 1 while all others are zero. By definition, the surfer's distribution at $t = 1$ is given by the probability vector $\vec{x}P$; at $t = 2$ by $(\vec{x}P)P = \vec{x}P^2$, and so on. We will detail this process in Section 21.2.2. We can thus compute the surfer's distribution over the states at any time, given only the initial distribution and the transition probability matrix P .

If a Markov chain is allowed to run for many time steps, each state is visited at a (different) frequency that depends on the structure of the Markov chain. In our running analogy, the surfer visits certain web pages (say, popular news home pages) more often than other pages. We now make this intuition precise, establishing conditions under which such the visit frequency converges to fixed, steady-state quantity. Following this, we set the Page-Rank of each node v to this steady-state visit frequency and show how it can be computed.

ERGODIC MARKOV CHAIN

Definition: A Markov chain is said to be *ergodic* if there exists a positive integer T_0 such that for all pairs of states i, j in the Markov chain, if it is started at time 0 in state i then for all $t > T_0$, the probability of being in state j at time t is greater than 0.

For a Markov chain to be ergodic, two technical conditions are required of its states and the non-zero transition probabilities; these conditions are known as *irreducibility* and *aperiodicity*. Informally, the first ensures that there is a sequence of transitions of non-zero probability from any state to any other, while the latter ensures that the states are not partitioned into sets such that all state transitions occur cyclically from one set to another.

STEADY-STATE

Theorem 21.1. *For any ergodic Markov chain, there is a unique steady-state probability vector $\vec{\pi}$ that is the principal left eigenvector of P , such that if $\eta(i, t)$ is the number of visits to state i in t steps, then*

$$\lim_{t \rightarrow \infty} \frac{\eta(i, t)}{t} = \pi(i),$$

where $\pi(i) > 0$ is the steady-state probability for state i .

It follows from Theorem 21.1 that the random walk with teleporting results in a unique distribution of steady-state probabilities over the states of the induced Markov chain. This steady-state probability for a state is the PageRank of the corresponding web page.

21.2.2 The PageRank computation

How do we compute PageRank values? Recall the definition of a left eigenvector from Equation 18.2; the left eigenvectors of the transition probability matrix P are N -vectors $\vec{\pi}$ such that

$$(21.2) \quad \vec{\pi} P = \lambda \vec{\pi}.$$

The N entries in the principal eigenvector $\vec{\pi}$ are the steady-state probabilities of the random walk with teleporting, and thus the PageRank values for the corresponding web pages. We may interpret Equation (21.2) as follows: if $\vec{\pi}$ is the probability distribution of the surfer across the web pages, he remains in the steady-state distribution $\vec{\pi}$. Given that $\vec{\pi}$ is the steady-state distribution, we have that $\pi P = 1\pi$, so 1 is an eigenvalue of P . Thus if we were to compute the principal left eigenvector of the matrix P — the one with eigenvalue 1 — we would have computed the PageRank values.

There are many algorithms available for computing left eigenvectors; the references at the end of Chapter 18 and the present chapter are a guide to these. We give here a rather elementary method, sometimes known as *power iteration*. If \vec{x} is the initial distribution over the states, then the distribution at time t is $\vec{x}P^t$. As t grows large, we would expect that the distribution $\vec{x}P^t$ ² is very similar to the distribution $\vec{x}P^{t+1}$, since for large t we would expect the Markov chain to attain its steady state. By Theorem 21.1 this is independent of the initial distribution \vec{x} . The power iteration method simulates the surfer's walk: begin at a state and run the walk for a large number of steps t , keeping track of the visit frequencies for each of the states. After a large number of steps t , these frequencies "settle down" so that the variation in the computed frequencies is below some predetermined threshold. We declare these tabulated frequencies to be the PageRank values.

We consider the web graph in Exercise 21.6 with $\alpha = 0.5$. The transition probability matrix of the surfer's walk with teleportation is then

$$(21.3) \quad P = \begin{pmatrix} 1/6 & 2/3 & 1/6 \\ 5/12 & 1/6 & 5/12 \\ 1/6 & 2/3 & 1/6 \end{pmatrix}.$$

Imagine that the surfer starts in state 1, corresponding to the initial probability distribution vector $\vec{x}_0 = (1 \ 0 \ 0)$. Then, after one step the distribution is

$$(21.4) \quad \vec{x}_0 P = (1/6 \ 2/3 \ 1/6) = \vec{x}_1.$$

2. Note that P^t represents P raised to the t th power, not the transpose of P which is denoted P^T .

\vec{x}_0	1	0	0
\vec{x}_1	1/6	2/3	1/6
\vec{x}_2	1/3	1/3	1/3
\vec{x}_3	1/4	1/2	1/4
\vec{x}_4	7/24	5/12	7/24
...
\vec{x}	5/18	4/9	5/18

► **Figure 21.3** The sequence of probability vectors.

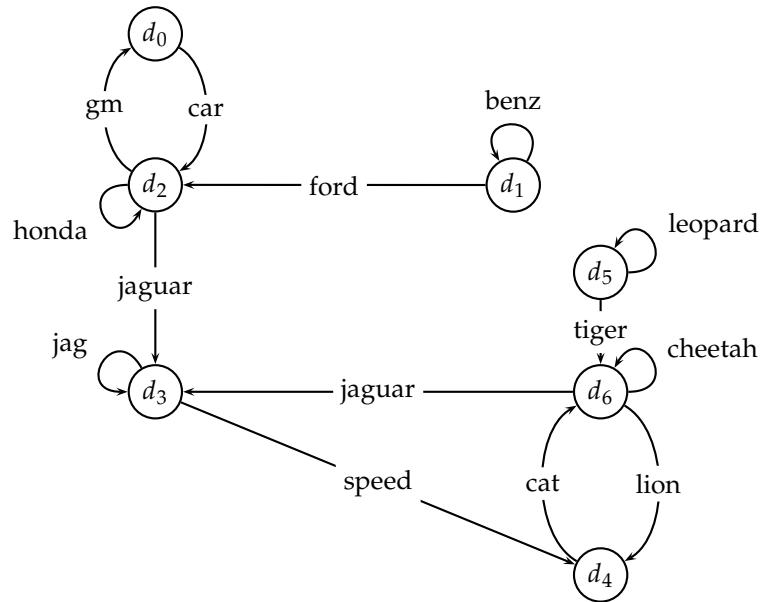
After two steps it is

$$(21.5) \quad \vec{x}_1 P = \begin{pmatrix} 1/6 & 2/3 & 1/6 \end{pmatrix} \begin{pmatrix} 1/6 & 2/3 & 1/6 \\ 5/12 & 1/6 & 5/12 \\ 1/6 & 2/3 & 1/6 \end{pmatrix} = \begin{pmatrix} 1/3 & 1/3 & 1/3 \end{pmatrix} = \vec{x}_2.$$

Continuing in this fashion gives a sequence of probability vectors as shown in Figure 21.3.

Continuing for several steps, we see that the distribution converges to the steady state of $\vec{x} = (5/18 \ 4/9 \ 5/18)$. In this simple example, we may directly calculate this steady-state probability distribution by observing the symmetry of the Markov chain: states 1 and 3 are symmetric, as evident from the fact that the first and third rows of the transition probability matrix in Equation (21.3) are identical. Postulating, then, that they both have the same steady-state probability and denoting this probability by p , we know that the steady-state distribution is of the form $\vec{\pi} = (p \ 1 - 2p \ p)$. Now, using the identity $\vec{\pi} = \vec{\pi}P$, we solve a simple linear equation to obtain $p = 5/18$ and consequently, $\vec{\pi} = (5/18 \ 4/9 \ 5/18)$.

The PageRank values of pages (and the implicit ordering amongst them) are independent of any query a user might pose; PageRank is thus a query-independent measure of the static quality of each web page (recall such static quality measures from Section 7.1.4). On the other hand, the relative ordering of pages should, intuitively, depend on the query being served. For this reason, search engines use static quality measures such as PageRank as just one of many factors in scoring a web page on a query. Indeed, the relative contribution of PageRank to the overall score may again be determined by machine-learned scoring as in Section 15.4.1.



► **Figure 21.4** A small web graph. Arcs are annotated with the word that occurs in the anchor text of the corresponding link.

 **Example 21.1:** Consider the graph in Figure 21.4. For a teleportation rate of 0.14 its (stochastic) transition probability matrix is:

$$\begin{matrix}
 0.02 & 0.02 & 0.88 & 0.02 & 0.02 & 0.02 & 0.02 \\
 0.02 & 0.45 & 0.45 & 0.02 & 0.02 & 0.02 & 0.02 \\
 0.31 & 0.02 & 0.31 & 0.31 & 0.02 & 0.02 & 0.02 \\
 0.02 & 0.02 & 0.02 & 0.45 & 0.45 & 0.02 & 0.02 \\
 0.02 & 0.02 & 0.02 & 0.02 & 0.02 & 0.02 & 0.88 \\
 0.02 & 0.02 & 0.02 & 0.02 & 0.02 & 0.45 & 0.45 \\
 0.02 & 0.02 & 0.02 & 0.31 & 0.31 & 0.02 & 0.31
 \end{matrix}$$

The PageRank vector of this matrix is:

$$(21.6) \quad \vec{x} = (0.05 \quad 0.04 \quad 0.11 \quad 0.25 \quad 0.21 \quad 0.04 \quad 0.31)$$

Observe that in Figure 21.4, q_2 , q_3 , q_4 and q_6 are the nodes with at least two in-links. Of these, q_2 has the lowest PageRank since the random walk tends to drift out of the top part of the graph – the walker can only return there through teleportation.

21.2.3 Topic-specific PageRank

Thus far we have discussed the PageRank computation with a teleport operation in which the surfer jumps to a random web page chosen uniformly at random. We now consider teleporting to a random web page chosen *non-uniformly*. In doing so, we are able to derive PageRank values tailored to particular interests. For instance, a sports aficionado might wish that pages on sports be ranked higher than non-sports pages. Suppose that web pages on sports are “near” one another in the web graph. Then, a random surfer who frequently finds himself on random sports pages is likely (in the course of the random walk) to spend most of his time at sports pages, so that the steady-state distribution of sports pages is boosted.

Suppose our random surfer, endowed with a teleport operation as before, teleports to *a random web page on the topic of sports* instead of teleporting to a uniformly chosen random web page. We will not focus on how we collect all web pages on the topic of sports; in fact, we only need a non-zero subset S of sports-related web pages, so that the teleport operation is feasible. This may be obtained, for instance, from a manually built directory of sports pages such as the open directory project (<http://www.dmoz.org/>) or that of Yahoo.

Provided the set S of sports-related pages is non-empty, it follows that there is a non-empty set of web pages $Y \supseteq S$ over which the random walk has a steady-state distribution; let us denote this *sports PageRank* distribution by $\vec{\pi}_s$. For web pages not in Y , we set the PageRank values to zero. We call $\vec{\pi}_s$ the *topic-specific PageRank* for sports.

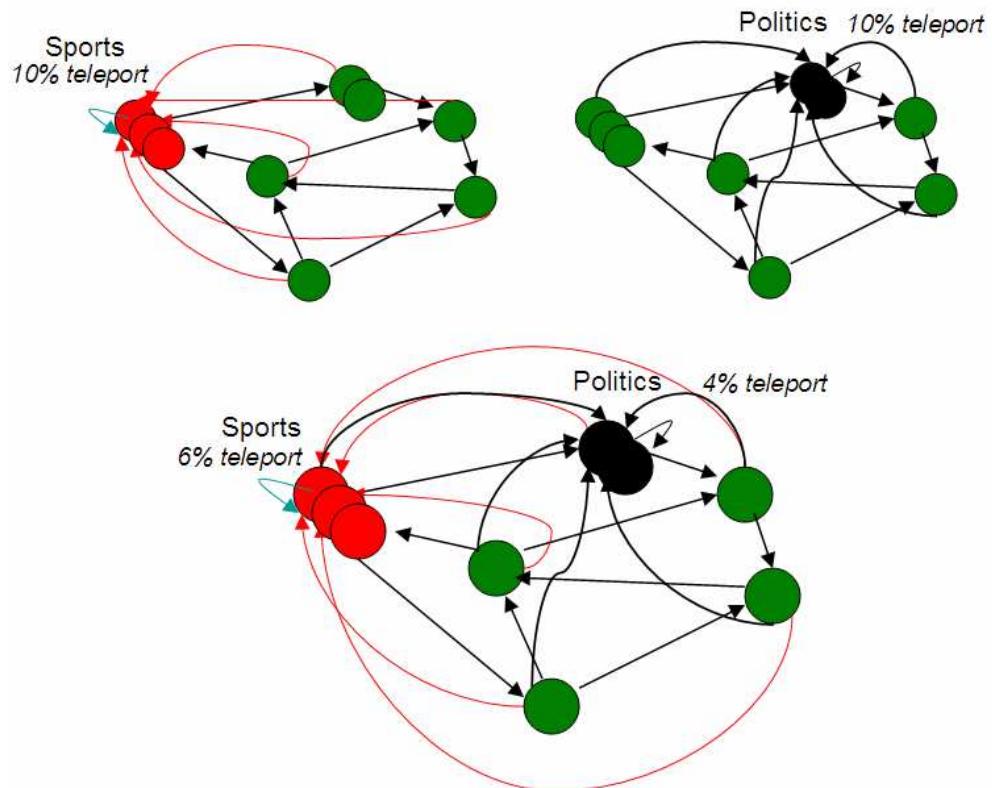
We do not demand that teleporting takes the random surfer to a uniformly chosen sports page; the distribution over teleporting targets S could in fact be arbitrary.

In like manner we can envision topic-specific PageRank distributions for each of several topics such as science, religion, politics and so on. Each of these distributions assigns to each web page a PageRank value in the interval $[0, 1]$. For a user interested in only a single topic from among these topics, we may invoke the corresponding PageRank distribution when scoring and ranking search results. This gives us the potential of considering settings in which the search engine knows what topic a user is interested in. This may happen because users either explicitly register their interests, or because the system learns by observing each user’s behavior over time.

But what if a user is known to have a mixture of interests from multiple topics? For instance, a user may have an interest mixture (or *profile*) that is 60% sports and 40% politics; can we compute a *personalized PageRank* for this user? At first glance, this appears daunting: how could we possibly compute a different PageRank distribution for each user profile (with, potentially, infinitely many possible profiles)? We can in fact address this provided we assume that an individual’s interests can be well-approximated as a linear

TOPIC-SPECIFIC
PAGERANK

PERSONALIZED
PAGERANK



► **Figure 21.5** Topic-specific PageRank. In this example we consider a user whose interests are 60% sports and 40% politics. If the teleportation probability is 10%, this user is modeled as teleporting 6% to sports pages and 4% to politics pages.

combination of a small number of topic page distributions. A user with this mixture of interests could teleport as follows: determine first whether to teleport to the set S of known sports pages, or to the set of known politics pages. This choice is made at random, choosing sports pages 60% of the time and politics pages 40% of the time. Once we choose that a particular teleport step is to (say) a random sports page, we choose a web page in S uniformly at random to teleport to. This in turn leads to an ergodic Markov chain with a steady-state distribution that is personalized to this user's preferences over topics (see Exercise 21.16).

While this idea has intuitive appeal, its implementation appears cumbersome: it seems to demand that for each user, we compute a transition prob-

ability matrix and compute its steady-state distribution. We are rescued by the fact that the evolution of the probability distribution over the states of a Markov chain can be viewed as a linear system. In Exercise 21.16 we will show that it is not necessary to compute a PageRank vector for every distinct combination of user interests over topics; the personalized PageRank vector for any user can be expressed as a linear combination of the underlying topic-specific PageRanks. For instance, the personalized PageRank vector for the user whose interests are 60% sports and 40% politics can be computed as

$$(21.7) \quad 0.6\vec{\pi}_s + 0.4\vec{\pi}_p,$$

where $\vec{\pi}_s$ and $\vec{\pi}_p$ are the topic-specific PageRank vectors for sports and for politics, respectively.



Exercise 21.5

Write down the transition probability matrix for the example in Figure 21.2.

Exercise 21.6

Consider a web graph with three nodes 1, 2 and 3. The links are as follows: $1 \rightarrow 2, 3 \rightarrow 2, 2 \rightarrow 1, 2 \rightarrow 3$. Write down the transition probability matrices for the surfer's walk with teleporting, for the following three values of the teleport probability: (a) $\alpha = 0$; (b) $\alpha = 0.5$ and (c) $\alpha = 1$.

Exercise 21.7

A user of a browser can, in addition to clicking a hyperlink on the page x he is currently browsing, use the *back button* to go back to the page from which he arrived at x . Can such a user of back buttons be modeled as a Markov chain? How would we model repeated invocations of the back button?

Exercise 21.8

Consider a Markov chain with three states A, B and C, and transition probabilities as follows. From state A, the next state is B with probability 1. From B, the next state is either A with probability p_A , or state C with probability $1 - p_A$. From C the next state is A with probability 1. For what values of $p_A \in [0, 1]$ is this Markov chain ergodic?

Exercise 21.9

Show that for any directed graph, the Markov chain induced by a random walk with the teleport operation is ergodic.

Exercise 21.10

Show that the PageRank of every page is at least α/N . What does this imply about the difference in PageRank values (over the various pages) as α becomes close to 1?

Exercise 21.11

For the data in Example 21.1, write a small routine or use a scientific calculator to compute the PageRank values stated in Equation (21.6).

Exercise 21.12

Suppose that the web graph is stored on disk as an adjacency list, in such a way that you may only query for the out-neighbors of pages in the order in which they are stored. You cannot load the graph in main memory but you may do multiple reads over the full graph. Write the algorithm for computing the PageRank in this setting.

Exercise 21.13

Recall the sets S and Y introduced near the beginning of Section 21.2.3. How does the set Y relate to S ?

Exercise 21.14

Is the set Y always the set of all web pages? Why or why not?

Exercise 21.15

[***]

Is the sports PageRank of any page in S at least as large as its PageRank?

Exercise 21.16

[***]

Consider a setting where we have two topic-specific PageRank values for each web page: a sports PageRank $\vec{\pi}_s$, and a politics PageRank $\vec{\pi}_p$. Let α be the (common) teleportation probability used in computing both sets of topic-specific PageRanks. For $q \in [0, 1]$, consider a user whose interest profile is divided between a fraction q in sports and a fraction $1 - q$ in politics. Show that the user's personalized PageRank is the steady-state distribution of a random walk in which – on a teleport step – the walk teleports to a sports page with probability q and to a politics page with probability $1 - q$.

Exercise 21.17

Show that the Markov chain corresponding to the walk in Exercise 21.16 is ergodic and hence the user's personalized PageRank can be obtained by computing the steady-state distribution of this Markov chain.

Exercise 21.18

Show that in the steady-state distribution of Exercise 21.17, the steady-state probability for any web page i equals $q\pi_s(i) + (1 - q)\pi_p(i)$.

21.3 Hubs and Authorities

HUB SCORE
AUTHORITY SCORE

We now develop a scheme in which, given a query, every web page is assigned *two* scores. One is called its *hub score* and the other its *authority score*. For any query, we compute two ranked lists of results rather than one. The ranking of one list is induced by the hub scores and that of the other by the authority scores.

This approach stems from a particular insight into the creation of web pages, that there are two primary kinds of web pages useful as results for *broad-topic searches*. By a broad topic search we mean an informational query such as "I wish to learn about leukemia". There are authoritative sources of information on the topic; in this case, the National Cancer Institute's page on

leukemia would be such a page. We will call such pages *authorities*; in the computation we are about to describe, they are the pages that will emerge with high authority scores.

On the other hand, there are many pages on the Web that are hand-compiled lists of links to authoritative web pages on a specific topic. These *hub* pages are not in themselves authoritative sources of topic-specific information, but rather compilations that someone with an interest in the topic has spent time putting together. The approach we will take, then, is to use these hub pages to discover the authority pages. In the computation we now develop, these hub pages are the pages that will emerge with high hub scores.

A good hub page is one that points to many good authorities; a good authority page is one that is pointed to by many good hub pages. We thus appear to have a circular definition of hubs and authorities; we will turn this into an iterative computation. Suppose that we have a subset of the web containing good hub and authority pages, together with the hyperlinks amongst them. We will iteratively compute a hub score and an authority score for every web page in this subset, deferring the discussion of how we pick this subset until Section 21.3.1.

For a web page v in our subset of the web, we use $h(v)$ to denote its hub score and $a(v)$ its authority score. Initially, we set $h(v) = a(v) = 1$ for all nodes v . We also denote by $v \mapsto y$ the existence of a hyperlink from v to y . The core of the iterative algorithm is a pair of updates to the hub and authority scores of all pages given by Equation 21.8, which capture the intuitive notions that good hubs point to good authorities and that good authorities are pointed to by good hubs.

$$(21.8) \quad \begin{aligned} h(v) &\leftarrow \sum_{v \mapsto y} a(y) \\ a(v) &\leftarrow \sum_{y \mapsto v} h(y). \end{aligned}$$

Thus, the first line of Equation (21.8) sets the hub score of page v to the sum of the authority scores of the pages it links to. In other words, if v links to pages with high authority scores, its hub score increases. The second line plays the reverse role; if page v is linked to by good hubs, its authority score increases.

What happens as we perform these updates iteratively, recomputing hub scores, then new authority scores based on the recomputed hub scores, and so on? Let us recast the equations Equation (21.8) into matrix-vector form. Let \vec{h} and \vec{a} denote the vectors of all hub and all authority scores respectively, for the pages in our subset of the web graph. Let A denote the adjacency matrix of the subset of the web graph that we are dealing with: A is a square matrix with one row and one column for each page in the subset. The entry

A_{ij} is 1 if there is a hyperlink from page i to page j , and 0 otherwise. Then, we may write Equation (21.8)

$$(21.9) \quad \begin{aligned} \vec{h} &\leftarrow A\vec{a} \\ \vec{a} &\leftarrow A^T\vec{h}, \end{aligned}$$

where A^T denotes the transpose of the matrix A . Now the right hand side of each line of Equation (21.9) is a vector that is the left hand side of the other line of Equation (21.9). Substituting these into one another, we may rewrite Equation (21.9) as

$$(21.10) \quad \begin{aligned} \vec{h} &\leftarrow AA^T\vec{h} \\ \vec{a} &\leftarrow A^TA\vec{a}. \end{aligned}$$

Now, Equation (21.10) bears an uncanny resemblance to a pair of eigenvector equations (Section 18.1); indeed, if we replace the \leftarrow symbols by $=$ symbols and introduce the (unknown) eigenvalue, the first line of Equation (21.10) becomes the equation for the eigenvectors of AA^T , while the second becomes the equation for the eigenvectors of A^TA :

$$(21.11) \quad \begin{aligned} \vec{h} &= (1/\lambda_h)AA^T\vec{h} \\ \vec{a} &= (1/\lambda_a)A^TA\vec{a}. \end{aligned}$$

Here we have used λ_h to denote the eigenvalue of AA^T and λ_a to denote the eigenvalue of A^TA .

This leads to some key consequences:

1. The iterative updates in Equation (21.8) (or equivalently, Equation (21.9)), if scaled by the appropriate eigenvalues, are equivalent to the power iteration method for computing the eigenvectors of AA^T and A^TA . Provided that the principal eigenvalue of AA^T is unique, the iteratively computed entries of \vec{h} and \vec{a} settle into unique steady-state values determined by the entries of A and hence the link structure of the graph.
2. In computing these eigenvector entries, we are not restricted to using the power iteration method; indeed, we could use any fast method for computing the principal eigenvector of a stochastic matrix.

The resulting computation thus takes the following form:

1. Assemble the target subset of web pages, form the graph induced by their hyperlinks and compute AA^T and A^TA .
2. Compute the principal eigenvectors of AA^T and A^TA to form the vector of hub scores \vec{h} and authority scores \vec{a} .

3. Output the top-scoring hubs and the top-scoring authorities.

HITS This method of link analysis is known as *HITS*, which is an acronym for *Hyperlink-Induced Topic Search*.



Example 21.2: Assuming the query *jaguar* and double-weighting of links whose anchors contain the query word, the matrix A for Figure 21.4 is as follows:

$$\begin{matrix} 0 & 0 & 1 & 0 & 0 & 0 & 0 \\ 0 & 1 & 1 & 0 & 0 & 0 & 0 \\ 1 & 0 & 1 & 2 & 0 & 0 & 0 \\ 0 & 0 & 0 & 1 & 1 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 1 \\ 0 & 0 & 0 & 0 & 0 & 1 & 1 \\ 0 & 0 & 0 & 2 & 1 & 0 & 1 \end{matrix}$$

The hub and authority vectors are:

$$\vec{h} = (0.03 \quad 0.04 \quad 0.33 \quad 0.18 \quad 0.04 \quad 0.04 \quad 0.35)$$

$$\vec{a} = (0.10 \quad 0.01 \quad 0.12 \quad 0.47 \quad 0.16 \quad 0.01 \quad 0.13)$$

Here, q_3 is the main authority – two hubs (q_2 and q_6) are pointing to it via highly weighted *jaguar* links.

Since the iterative updates captured the intuition of good hubs and good authorities, the high-scoring pages we output would give us good hubs and authorities from the target subset of web pages. In Section 21.3.1 we describe the remaining detail: how do we gather a target subset of web pages around a topic such as leukemia?

21.3.1 Choosing the subset of the Web

In assembling a subset of web pages around a topic such as leukemia, we must cope with the fact that good authority pages may not contain the specific query term leukemia. This is especially true, as we noted in Section 21.1.1, when an authority page uses its web presence to project a certain marketing image. For instance, many pages on the IBM website are authoritative sources of information on computer hardware, even though these pages may not contain the term computer or hardware. However, a hub compiling computer hardware resources is likely to use these terms and also link to the relevant pages on the IBM website.

Building on these observations, the following procedure has been suggested for compiling the subset of the Web for which to compute hub and authority scores.

1. Given a query (say leukemia), use a text index to get all pages containing leukemia. Call this the *root set* of pages.
2. Build the *base set* of pages, to include the root set as well as any page that either links to a page in the root set, or is linked to by a page in the root set.

We then use the base set for computing hub and authority scores. The base set is constructed in this manner for three reasons:

1. A good authority page may not contain the query text (such as computer hardware).
2. If the text query manages to capture a good hub page v_h in the root set, then the inclusion of all pages linked to by any page in the root set will capture all the good authorities linked to by v_h in the base set.
3. Conversely, if the text query manages to capture a good authority page v_a in the root set, then the inclusion of pages which point to v_a will bring other good hubs into the base set. In other words, the “expansion” of the root set into the base set enriches the common pool of good hubs and authorities.

Running HITS across a variety of queries reveals some interesting insights about link analysis. Frequently, the documents that emerge as top hubs and authorities include languages other than the language of the query. These pages were presumably drawn into the base set, following the assembly of the root set. Thus, some elements of *cross-language retrieval* (where a query in one language retrieves documents in another) are evident here; interestingly, this cross-language effect resulted purely from link analysis, with no linguistic translation taking place.

We conclude this section with some notes on implementing this algorithm. The root set consists of all pages matching the text query; in fact, implementations (see the references in Section 21.4) suggest that it suffices to use 200 or so web pages for the root set, rather than all pages matching the text query. Any algorithm for computing eigenvectors may be used for computing the hub/authority score vector. In fact, we need not compute the exact values of these scores; it suffices to know the relative values of the scores so that we may identify the top hubs and authorities. To this end, it is possible that a small number of iterations of the power iteration method yields the relative ordering of the top hubs and authorities. Experiments have suggested that in practice, about five iterations of Equation (21.8) yield fairly good results. Moreover, since the link structure of the web graph is fairly sparse (the average web page links to about ten others), we do not perform these as matrix-vector products but rather as additive updates as in Equation (21.8).

► **Figure 21.6** A sample run of HITS on the query japan elementary schools.

Figure 21.6 shows the results of running HITS on the query japan elementary schools. The figure shows the top hubs and authorities; each row lists the title tag from the corresponding HTML page. Because the resulting string is not necessarily in Latin characters, the resulting print is (in many cases) a string of gibberish. Each of these corresponds to a web page that does not use Latin characters, in this case very likely pages in Japanese. There also appear to be pages in other non-English languages, which seems surprising given that the query string is in English. In fact, this result is emblematic of the functioning of HITS – following the assembly of the root set, the (English) query string is ignored. The base set is likely to contain pages in other languages, for instance if an English-language hub page links to the Japanese-language home pages of Japanese elementary schools. Because the subsequent computation of the top hubs and authorities is entirely link-based, some of these non-English pages will appear among the top hubs and authorities.

Exercise 21.19

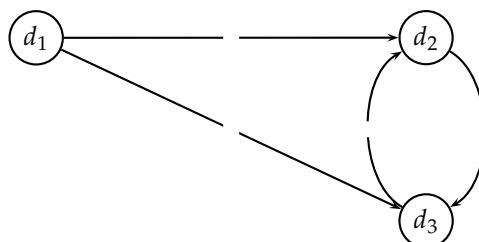
- If all the hub and authority scores are initialized to 1, what is the hub/authority score of a node after one iteration?

Exercise 21.20

How would you interpret the entries of the matrices AA^T and A^TA ? What is the connection to the co-occurrence matrix CC^T in Chapter 18?

Exercise 21.21

What are the principal eigenvalues of AA^T and A^TA ?



► **Figure 21.7** Web graph for Exercise 21.22.

Exercise 21.22

For the web graph in Figure 21.7, compute PageRank, hub and authority scores for each of the three pages. Also give the relative ordering of the 3 nodes for each of these scores, indicating any ties.

PageRank: Assume that at each step of the PageRank random walk, we teleport to a random page with probability 0.1, with a uniform distribution over which particular page we teleport to.

Hubs/Authorities: Normalize the hub (authority) scores so that the maximum hub (authority) score is 1.

Hint 1: Using symmetries to simplify and solving with linear equations might be easier than using iterative methods.

Hint 2: Provide the relative ordering (indicating any ties) of the three nodes for each of the three scoring measures.

21.4 References and further reading

Garfield (1955) is seminal in the science of citation analysis. This was built on by **Pinski and Narin (1976)** to develop a *journal influence weight*, whose definition is remarkably similar to that of the PageRank measure.

The use of anchor text as an aid to searching and ranking stems from the work of **McBryan (1994)**. Extended anchor-text was implicit in his work, with systematic experiments reported in **Chakrabarti et al. (1998)**.

Kemeny and Snell (1976) is a classic text on Markov chains. The PageRank measure was developed in **Brin and Page (1998)** and in **Page et al. (1998)**.

A number of methods for the fast computation of PageRank values are surveyed in Berkhin (2005) and in Langville and Meyer (2006); the former also details how the PageRank eigenvector solution may be viewed as solving a linear system, leading to one way of solving Exercise 21.16. The effect of the teleport probability α has been studied by Baeza-Yates et al. (2005) and by Boldi et al. (2005). Topic-specific PageRank and variants were developed in Haveliwala (2002), Haveliwala (2003) and in Jeh and Widom (2003). Berkhin (2006a) develops an alternate view of topic-specific PageRank.

LINK FARMS

Ng et al. (2001b) suggests that the PageRank score assignment is more robust than HITS in the sense that scores are less sensitive to small changes in graph topology. However, it has also been noted that the teleport operation contributes significantly to PageRank's robustness in this sense. Both PageRank and HITS can be "spammed" by the orchestrated insertion of links into the web graph; indeed, the Web is known to have such *link farms* that collude to increase the score assigned to certain pages by various link analysis algorithms.

The HITS algorithm is due to Kleinberg (1999). Chakrabarti et al. (1998) developed variants that weighted links in the iterative computation based on the presence of query terms in the pages being linked and compared these to results from several web search engines. Bharat and Henzinger (1998) further developed these and other heuristics, showing that certain combinations outperformed the basic HITS algorithm. Borodin et al. (2001) provides a systematic study of several variants of the HITS algorithm. Ng et al. (2001b) introduces a notion of *stability* for link analysis, arguing that small changes to link topology should not lead to significant changes in the ranked list of results for a query. Numerous other variants of HITS have been developed by a number of authors, the best known of which is perhaps SALSA (Lempel and Moran 2000).