Beej's Guide to Network Programming Using Internet Sockets

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1. Intro

Hey! Socket programming got you down? Is this stuff just a little too difficult to figure out from the **man** pages? You want to do cool Internet programming, but you don't have time to wade through a gob of structs trying to figure out if you have to call bind() before you connect(), etc., etc.

Well, guess what! I've already done this nasty business, and I'm dying to share the information with everyone! You've come to the right place. This document should give the average competent C programmer the edge s/he needs to get a grip on this networking noise.

1.1. Audience

This document has been written as a tutorial, not a reference. It is probably at its best when read by individuals who are just starting out with socket programming and are looking for a foothold. It is certainly not the *complete* guide to sockets programming, by any means.

Hopefully, though, it'll be just enough for those man pages to start making sense... :-)

1.2. Platform and Compiler

The code contained within this document was compiled on a Linux PC using Gnu's **gcc** compiler. It should, however, build on just about any platform that uses **gcc**. Naturally, this doesn't apply if you're programming for Windows–see the section on Windows programming, below.

1.3. Official Homepage

This official location of this document is http://beej.us/guide/bgnet/.

1.4. Note for Solaris/SunOS Programmers

When compiling for Solaris or SunOS, you need to specify some extra command-line switches for linking in the proper libraries. In order to do this, simply add "-lnsl -lsocket-lresolv" to the end of the compile command, like so:

```
$ cc -o server server.c -lnsl -lsocket -lresolv
```

If you still get errors, you could try further adding a "-lxnet" to the end of that command line. I don't know what that does, exactly, but some people seem to need it.

Another place that you might find problems is in the call to setsockopt(). The prototype differs from that on my Linux box, so instead of:

```
int yes=1;
enter this:
char yes='1';
```

As I don't have a Sun box, I haven't tested any of the above information—it's just what people have told me through email.

1.5. Note for Windows Programmers

I have a particular dislike for Windows, and encourage you to try Linux, BSD, or Unix instead. That being said, you can still use this stuff under Windows.

First, ignore pretty much all of the system header files I mention in here. All you need to include is:

```
#include <winsock.h>
```

Wait! You also have to make a call to WSAStartup() before doing anything else with the sockets library. The code to do that looks something like this:

You also have to tell your compiler to link in the Winsock library, usually called **wsock32.lib** or **winsock32.lib** or somesuch. Under VC++, this can be done through the Project menu, under Settings.... Click the Link tab, and look for the box titled "Object/library modules". Add "wsock32.lib" to that list.

Or so I hear.

Finally, you need to call WSACleanup() when you're all through with the sockets library. See your online help for details.

Once you do that, the rest of the examples in this tutorial should generally apply, with a few exceptions. For one thing, you can't use close() to close a socket—you need to use closesocket(), instead. Also, select() only works with socket descriptors, not file descriptors (like 0 for stdin).

There is also a socket class that you can use, CSocket. Check your compilers help pages for more information.

To get more information about Winsock, read the Winsock FAQ¹ and go from there.

Finally, I hear that Windows has no fork() system call which is, unfortunately, used in some of my examples. Maybe you have to link in a POSIX library or something to get it to work, or you can use CreateProcess() instead. fork() takes no arguments, and CreateProcess() takes about 48 billion arguments. If you're not up to that, the CreateThread() is a little easier to digest...unfortunately a discussion about multithreading is beyond the scope of this document. I can only talk about so much, you know!

1.6. Email Policy

I'm generally available to help out with email questions so feel free to write in, but I can't guarantee a response. I lead a pretty busy life and there are times when I just can't answer a question you have. When that's the case, I usually just delete the message. It's nothing personal; I just won't ever have the time to give the detailed answer you require.

As a rule, the more complex the question, the less likely I am to respond. If you can narrow down your question before mailing it and be sure to include any pertinent information (like platform, compiler, error messages you're getting, and anything else you think might help me troubleshoot), you're much more likely to get a response. For more pointers, read ESR's document, How To Ask Questions The Smart Way².

If you don't get a response, hack on it some more, try to find the answer, and if it's still elusive, then write me again with the information you've found and hopefully it will be enough for me to help out.

¹ http://tangentsoft.net/wskfaq/

² http://www.catb.org/~esr/faqs/smart-questions.html

Now that I've badgered you about how to write and not write me, I'd just like to let you know that I *fully* appreciate all the praise the guide has received over the years. It's a real morale boost, and it gladdens me to hear that it is being used for good! :-) Thank you!

1.7. Mirroring

You are more than welcome to mirror this site, whether publically or privately. If you publically mirror the site and want me to link to it from the main page, drop me a line at beej@beej.us.

1.8. Note for Translators

If you want to translate the guide into another language, write me at beej@beej.us and I'll link to your translation from the main page.

Feel free to add your name and email address to the translation.

Sorry, but due to space constraints, I cannot host the translations myself.

1.9. Copyright and Distribution

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The C source code presented in this document is hereby granted to the public domain. Contact *beej@beej.us* for more information.

2. What is a socket?

You hear talk of "sockets" all the time, and perhaps you are wondering just what they are exactly. Well, they're this: a way to speak to other programs using standard Unix file descriptors. What?

Ok—you may have heard some Unix hacker state, "Jeez, *everything* in Unix is a file!" What that person may have been talking about is the fact that when Unix programs do any sort of I/O, they do it by reading or writing to a file descriptor. A file descriptor is simply an integer associated with an open file. But (and here's the catch), that file can be a network connection, a FIFO, a pipe, a terminal, a real on-the-disk file, or just about anything else. Everything in Unix *is* a file! So when you want to communicate with another program over the Internet you're gonna do it through a file descriptor, you'd better believe it.

"Where do I get this file descriptor for network communication, Mr. Smarty-Pants?" is probably the last question on your mind right now, but I'm going to answer it anyway: You make a call to the socket() system routine. It returns the socket descriptor, and you communicate through it using the specialized send() and recv() (man send³, man recv⁴) socket calls.

"But, hey!" you might be exclaiming right about now. "If it's a file descriptor, why in the name of Neptune can't I just use the normal read() and write() calls to communicate through the socket?" The short answer is, "You can!" The longer answer is, "You can, but send() and recv() offer much greater control over your data transmission."

What next? How about this: there are all kinds of sockets. There are DARPA Internet addresses (Internet Sockets), path names on a local node (Unix Sockets), CCITT X.25 addresses (X.25 Sockets that you can safely ignore), and probably many others depending on which Unix flavor you run. This document deals only with the first: Internet Sockets.

2.1. Two Types of Internet Sockets

What's this? There are two types of Internet sockets? Yes. Well, no. I'm lying. There are more, but I didn't want to scare you. I'm only going to talk about two types here. Except for this sentence, where I'm going to tell you that "Raw Sockets" are also very powerful and you should look them up.

All right, already. What are the two types? One is "Stream Sockets"; the other is "Datagram Sockets", which may hereafter be referred to as "SOCK_STREAM" and "SOCK_DGRAM", respectively. Datagram sockets are sometimes called "connectionless sockets". (Though they can be connect()'d if you really want. See connect(), below.)

Stream sockets are reliable two-way connected communication streams. If you output two items into the socket in the order "1, 2", they will arrive in the order "1, 2" at the opposite end. They will also be error free. Any errors you do encounter are figments of your own deranged mind, and are not to be discussed here.

What uses stream sockets? Well, you may have heard of the **telnet** application, yes? It uses stream sockets. All the characters you type need to arrive in the same order you type them, right? Also, web browsers use the HTTP protocol which uses stream sockets to get pages. Indeed, if you telnet to a web site on port 80, and type "GET / HTTP/1.0" and hit RETURN twice, it'll dump the HTML back at you!

How do stream sockets achieve this high level of data transmission quality? They use a protocol called "The Transmission Control Protocol", otherwise known as "TCP" (see RFC-793⁵ for extremely detailed info on TCP.) TCP makes sure your data arrives sequentially and

³ http://man.linuxquestions.org/index.php?query=send§ion=2&type=2

⁴ http://man.linuxquestions.org/index.php?query=recv§ion=2&type=2

⁵http://www.rfc-editor.org/rfc/rfc793.txt

error-free. You may have heard "TCP" before as the better half of "TCP/IP" where "IP" stands for "Internet Protocol" (see RFC-791⁶.) IP deals primarily with Internet routing and is not generally responsible for data integrity.

Cool. What about Datagram sockets? Why are they called connectionless? What is the deal, here, anyway? Why are they unreliable? Well, here are some facts: if you send a datagram, it may arrive. It may arrive out of order. If it arrives, the data within the packet will be error-free.

Datagram sockets also use IP for routing, but they don't use TCP; they use the "User Datagram Protocol", or "UDP" (see RFC-768⁷.)

Why are they connectionless? Well, basically, it's because you don't have to maintain an open connection as you do with stream sockets. You just build a packet, slap an IP header on it with destination information, and send it out. No connection needed. They are generally used for packet-by-packet transfers of information. Sample applications: **tftp**, **bootp**, etc.

"Enough!" you may scream. "How do these programs even work if datagrams might get lost?!" Well, my human friend, each has it's own protocol on top of UDP. For example, the tftp protocol says that for each packet that gets sent, the recipient has to send back a packet that says, "I got it!" (an "ACK" packet.) If the sender of the original packet gets no reply in, say, five seconds, he'll re-transmit the packet until he finally gets an ACK. This acknowledgment procedure is very important when implementing SOCK_DGRAM applications.

2.2. Low level Nonsense and Network Theory

Since I just mentioned layering of protocols, it's time to talk about how networks really work, and to show some examples of how SOCK_DGRAM packets are built. Practically, you can probably skip this section. It's good background, however.



Data Encapsulation.

Hey, kids, it's time to learn about *Data Encapsulation*! This is very very important. It's so important that you might just learn about it if you take the networks course here at Chico State;—). Basically, it says this: a packet is born, the packet is wrapped ("encapsulated") in a header (and rarely a footer) by the first protocol (say, the TFTP protocol), then the whole thing (TFTP header included) is encapsulated again by the next protocol (say, UDP), then again by the next (IP), then again by the final protocol on the hardware (physical) layer (say, Ethernet).

When another computer receives the packet, the hardware strips the Ethernet header, the kernel strips the IP and UDP headers, the TFTP program strips the TFTP header, and it finally has the data.

Now I can finally talk about the infamous *Layered Network Model*. This Network Model describes a system of network functionality that has many advantages over other models. For instance, you can write sockets programs that are exactly the same without caring how the data is physically transmitted (serial, thin Ethernet, AUI, whatever) because programs on lower levels deal with it for you. The actual network hardware and topology is transparent to the socket programmer.

Without any further ado, I'll present the layers of the full-blown model. Remember this for network class exams:

⁶ http://www.rfc-editor.org/rfc/rfc791.txt

⁷http://www.rfc-editor.org/rfc/rfc768.txt

- Application
- Presentation
- Session
- Transport
- Network
- Data Link
- Physical

The Physical Layer is the hardware (serial, Ethernet, etc.). The Application Layer is just about as far from the physical layer as you can imagine—it's the place where users interact with the network.

Now, this model is so general you could probably use it as an automobile repair guide if you really wanted to. A layered model more consistent with Unix might be:

- Application Layer (telnet, ftp, etc.)
- Host-to-Host Transport Layer (*TCP*, *UDP*)
- Internet Layer (*IP and routing*)
- Network Access Layer (Ethernet, ATM, or whatever)

At this point in time, you can probably see how these layers correspond to the encapsulation of the original data.

See how much work there is in building a simple packet? Jeez! And you have to type in the packet headers yourself using "cat"! Just kidding. All you have to do for stream sockets is send() the data out. All you have to do for datagram sockets is encapsulate the packet in the method of your choosing and sendto() it out. The kernel builds the Transport Layer and Internet Layer on for you and the hardware does the Network Access Layer. Ah, modern technology.

So ends our brief foray into network theory. Oh yes, I forgot to tell you everything I wanted to say about routing: nothing! That's right, I'm not going to talk about it at all. The router strips the packet to the IP header, consults its routing table, blah blah. Check out the IP RFC⁸ if you really really care. If you never learn about it, well, you'll live.

⁸ http://www.rfc-editor.org/rfc/rfc791.txt

3. structs and Data Handling

Well, we're finally here. It's time to talk about programming. In this section, I'll cover various data types used by the sockets interface, since some of them are a real bear to figure out. First the easy one: a socket descriptor. A socket descriptor is the following type:

```
int
```

Just a regular int.

Things get weird from here, so just read through and bear with me. Know this: there are two byte orderings: most significant byte (sometimes called an "octet") first, or least significant byte first. The former is called "Network Byte Order". Some machines store their numbers internally in Network Byte Order, some don't. When I say something has to be in Network Byte Order, you have to call a function (such as htons()) to change it from "Host Byte Order". If I don't say "Network Byte Order", then you must leave the value in Host Byte Order.

(For the curious, "Network Byte Order" is also known as "Big-Endian Byte Order".)

My First Struct TM -struct sockaddr. This structure holds socket address information for many types of sockets:

```
struct sockaddr {
   unsigned short sa_family; // address family, AF_xxx
   char sa_data[14]; // 14 bytes of protocol address
};
```

sa_family can be a variety of things, but it'll be AF_INET for everything we do in this document. sa_data contains a destination address and port number for the socket. This is rather unwieldy since you don't want to tediously pack the address in the sa_data by hand.

To deal with struct sockaddr, programmers created a parallel structure: struct sockaddr_in ("in" for "Internet".)

This structure makes it easy to reference elements of the socket address. Note that sin_zero (which is included to pad the structure to the length of a struct sockaddr) should be set to all zeros with the function memset(). Also, and this is the *important* bit, a pointer to a struct sockaddr_in can be cast to a pointer to a struct sockaddr and vice-versa. So even though connect() wants a struct sockaddr*, you can still use a struct sockaddr_in and cast it at the last minute! Also, notice that sin_family corresponds to sa_family in a struct sockaddr and should be set to "AF_INET". Finally, the sin_port and sin_addr must be in Network Byte Order!

"But," you object, "how can the entire structure, struct in_addr sin_addr, be in Network Byte Order?" This question requires careful examination of the structure struct in_addr, one of the worst unions alive:

```
// Internet address (a structure for historical reasons)
struct in_addr {
   unsigned long s_addr; // that's a 32-bit long, or 4 bytes
};
```

Well, it *used* to be a union, but now those days seem to be gone. Good riddance. So if you have declared *ina* to be of type struct sockaddr_in, then *ina.sin_addr.s_addr*

references the 4-byte IP address (in Network Byte Order). Note that even if your system still uses the God-awful union for struct in_addr, you can still reference the 4-byte IP address in exactly the same way as I did above (this due to #defines.)

3.1. Convert the Natives!

We've now been lead right into the next section. There's been too much talk about this Network to Host Byte Order conversion—now is the time for action!

All righty. There are two types that you can convert: short (two bytes) and long (four bytes). These functions work for the unsigned variations as well. Say you want to convert a short from Host Byte Order to Network Byte Order. Start with "h" for "host", follow it with "to", then "n" for "network", and "s" for "short": h-to-n-s, or htons() (read: "Host to Network Short").

It's almost too easy...

You can use every combination of "n", "h", "s", and "l" you want, not counting the really stupid ones. For example, there is NOT a stolh() ("Short to Long Host") function—not at this party, anyway. But there are:

- htons() "Host to Network Short"
- htonl() "Host to Network Long"
- ntohs() "Network to Host Short"
- ntohl() "Network to Host Long"

Now, you may think you're wising up to this. You might think, "What do I do if I have to change byte order on a char?" Then you might think, "Uh, never mind." You might also think that since your 68000 machine already uses network byte order, you don't have to call htonl() on your IP addresses. You would be right, *BUT* if you try to port to a machine that has reverse network byte order, your program will fail. Be portable! This is a Unix world! (As much as Bill Gates would like to think otherwise.) Remember: put your bytes in Network Byte Order before you put them on the network.

A final point: why do sin_addr and sin_port need to be in Network Byte Order in a struct $sockaddr_in$, but sin_family does not? The answer: sin_addr and sin_port get encapsulated in the packet at the IP and UDP layers, respectively. Thus, they must be in Network Byte Order. However, the sin_family field is only used by the kernel to determine what type of address the structure contains, so it must be in Host Byte Order. Also, since sin_family does not get sent out on the network, it can be in Host Byte Order.

3.2. IP Addresses and How to Deal With Them

Fortunately for you, there are a bunch of functions that allow you to manipulate IP addresses. No need to figure them out by hand and stuff them in a long with the << operator.

First, let's say you have a struct sockaddr_in ina, and you have an IP address "10.12.110.57" that you want to store into it. The function you want to use, inet_addr(), converts an IP address in numbers-and-dots notation into an unsigned long. The assignment can be made as follows:

```
ina.sin_addr.s_addr = inet_addr("10.12.110.57");
```

Notice that inet_addr() returns the address in Network Byte Order already—you don't have to call htonl(). Swell!

Now, the above code snippet isn't very robust because there is no error checking. See, inet_addr() returns -1 on error. Remember binary numbers? (unsigned)-1 just happens to correspond to the IP address 255.255.255! That's the broadcast address! Wrongo. Remember to do your error checking properly.

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Actually, there's a cleaner interface you can use instead of inet_addr(): it's called inet_aton() ("aton" means "ascii to network"):

```
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
int inet_aton(const char *cp, struct in_addr *inp);
```

And here's a sample usage, while packing a struct sockaddr_in (this example will make more sense to you when you get to the sections on bind() and connect().)

inet_aton(), *unlike practically every other socket-related function*, returns non-zero on success, and zero on failure. And the address is passed back in *inp*.

Unfortunately, not all platforms implement inet_aton() so, although its use is preferred, the older more common inet_addr() is used in this guide.

All right, now you can convert string IP addresses to their binary representations. What about the other way around? What if you have a struct in_addr and you want to print it in numbers-and-dots notation? In this case, you'll want to use the function inet_ntoa() ("ntoa" means "network to ascii") like this:

```
printf("%s", inet_ntoa(ina.sin_addr));
```

That will print the IP address. Note that inet_ntoa() takes a struct in_addr as an argument, not a long. Also notice that it returns a pointer to a char. This points to a statically stored char array within inet_ntoa() so that each time you call inet_ntoa() it will overwrite the last IP address you asked for. For example:

```
char *a1, *a2;
a1 = inet_ntoa(ina1.sin_addr);  // this is 192.168.4.14
a2 = inet_ntoa(ina2.sin_addr);  // this is 10.12.110.57
printf("address 1: %s\n",a1);
printf("address 2: %s\n",a2);

will print:

address 1: 10.12.110.57
address 2: 10.12.110.57
```

If you need to save the address, strcpy() it to your own character array.

That's all on this topic for now. Later, you'll learn to convert a string like "whitehouse.gov" into its corresponding IP address (see DNS, below.)

3.2.1: Private (Or Disconnected) Networks

Lots of places have a firewall that hides the network from the rest of the world for their own protection. And often times, the firewall translates "internal" IP addresses to "external" (that everyone else in the world knows) IP addresses using a process called *Network Address Translation*, or NAT.

Are you getting nervous yet? "Where's he going with all this weird stuff?"

Well, relax and buy yourself a drink, because as a beginner, you don't even have to worry about NAT, since it's done for you transparently. But I wanted to talk about the network behind the firewall in case you started getting confused by the network numbers you were seeing.

For instance, I have a firewall at home. I have two static IP addresses allocated to me by the DSL company, and yet I have seven computers on the network. How is this possible? Two computers can't share the same IP address, or else the data wouldn't know which one to go to!

The answer is: they don't share the same IP addresses. They are on a private network with 24 million IP addresses allocated to it. They are all just for me. Well, all for me as far as anyone else is concerned. Here's what's happening:

If I log into a remote computer, it tells me I'm logged in from 64.81.52.10 (not my real IP). But if I ask my local computer what it's IP address is, it says 10.0.0.5. Who is translating the IP address from one to the other? That's right, the firewall! It's doing NAT!

10.x.x.x is one of a few reserved networks that are only to be used either on fully disconnected networks, or on networks that are behind firewalls. The details of which private network numbers are available for you to use are outlined in RFC 19189, but some common ones you'll see are 10.x.x.x and 192.168.x.x, where x is 0-255, generally. Less common is 172.y.x.x, where y goes between 16 and 31.

Networks behind a NATing firewall don't need to be on one of these reserved networks, but they commonly are.

⁹ ftp://ftp.rfc-editor.org/in-notes/rfc1918.txt

4. System Calls or Bust

This is the section where we get into the system calls that allow you to access the network functionality of a Unix box. When you call one of these functions, the kernel takes over and does all the work for you automagically.

The place most people get stuck around here is what order to call these things in. In that, the man pages are no use, as you've probably discovered. Well, to help with that dreadful situation, I've tried to lay out the system calls in the following sections in exactly (approximately) the same order that you'll need to call them in your programs.

That, coupled with a few pieces of sample code here and there, some milk and cookies (which I fear you will have to supply yourself), and some raw guts and courage, and you'll be beaming data around the Internet like the Son of Jon Postel!

4.1. socket()—Get the File Descriptor!

I guess I can put it off no longer—I have to talk about the socket() system call. Here's the breakdown:

```
#include <sys/types.h>
#include <sys/socket.h>
int socket(int domain, int type, int protocol);
```

But what are these arguments? First, domain should be set to "PF_INET". Next, the type argument tells the kernel what kind of socket this is: SOCK_STREAM or SOCK_DGRAM. Finally, just set protocol to "0" to have socket() choose the correct protocol based on the type. (Notes: there are many more domains than I've listed. There are many more types than I've listed. See the socket() man page. Also, there's a "better" way to get the protocol, but specifying 0 works in 99.9% of all cases. See the getprotobyname() man page if you're curious.)

socket () simply returns to you a socket descriptor that you can use in later system calls, or -1 on error. The global variable errno is set to the error's value (see the perror() man page.)

(This PF_INET thing is a close relative of the AF_INET that you used when initializing the sin_family field in your struct sockaddr_in. In fact, they're so closely related that they actually have the same value, and many programmers will call socket() and pass AF_INET as the first argument instead of PF_INET. Now, get some milk and cookies, because it's times for a story. Once upon a time, a long time ago, it was thought that maybe a address family (what the "AF" in "AF_INET" stands for) might support several protocols that were referred to by their protocol family (what the "PF" in "PF_INET" stands for). That didn't happen. And they all lived happily ever after, The End. So the most correct thing to do is to use AF_INET in your struct sockaddr_in and PF_INET in your call to socket().)

Fine, fine, fine, but what good is this socket? The answer is that it's really no good by itself, and you need to read on and make more system calls for it to make any sense.

4.2. bind()—What port am I on?

Once you have a socket, you might have to associate that socket with a port on your local machine. (This is commonly done if you're going to listen() for incoming connections on a specific port–MUDs do this when they tell you to "telnet to x.y.z port 6969".) The port number is used by the kernel to match an incoming packet to a certain process's socket descriptor. If you're going to only be doing a connect(), this may be unnecessary. Read it anyway, just for kicks.

Here is the synopsis for the bind() system call:

```
#include <sys/types.h>
#include <sys/socket.h>
```

```
int bind(int sockfd, struct sockaddr *my_addr, int addrlen);
```

sockfd is the socket file descriptor returned by socket(). my_addr is a pointer to a struct sockaddr that contains information about your address, namely, port and IP address. addrlen can be set to size of (struct sockaddr).

Whew. That's a bit to absorb in one chunk. Let's have an example:

```
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#define MYPORT 3490
main()
    int sockfd;
    struct sockaddr_in my_addr;
    sockfd = socket(PF_INET, SOCK_STREAM, 0); // do some error checking!
    my_addr.sin_family = AF_INET;
                                          // host byte order
    my_addr.sin_port = htons(MYPORT);
                                          // short, network byte order
    my_addr.sin_addr.s_addr = inet_addr("10.12.110.57");
    memset(&(my_addr.sin_zero), '\0', 8); // zero the rest of the struct
    // don't forget your error checking for bind():
    bind(sockfd, (struct sockaddr *)&my_addr, sizeof(struct sockaddr));
```

There are a few things to notice here: my_addr.sin_port is in Network Byte Order. So is my_addr.sin_addr.s_addr. Another thing to watch out for is that the header files might differ from system to system. To be sure, you should check your local man pages.

Lastly, on the topic of bind(), I should mention that some of the process of getting your own IP address and/or port can be automated:

```
my_addr.sin_port = 0; // choose an unused port at random
my_addr.sin_addr.s_addr = INADDR_ANY; // use my IP address
```

See, by setting my_addr.sin_port to zero, you are telling bind() to choose the port for you. Likewise, by setting my_addr.sin_addr.s_addr to INADDR_ANY, you are telling it to automatically fill in the IP address of the machine the process is running on.

If you are into noticing little things, you might have seen that I didn't put INADDR_ANY into Network Byte Order! Naughty me. However, I have inside info: INADDR_ANY is really zero! Zero still has zero on bits even if you rearrange the bytes. However, purists will point out that there could be a parallel dimension where INADDR_ANY is, say, 12 and that my code won't work there. That's ok with me:

```
my_addr.sin_port = htons(0); // choose an unused port at random
my_addr.sin_addr.s_addr = htonl(INADDR_ANY); // use my IP address
```

Now we're so portable you probably wouldn't believe it. I just wanted to point that out, since most of the code you come across won't bother running INADDR_ANY through htonl().

bind() also returns -1 on error and sets errno to the error's value.

Another thing to watch out for when calling bind(): don't go underboard with your port numbers. All ports below 1024 are RESERVED (unless you're the superuser)! You can have any port number above that, right up to 65535 (provided they aren't already being used by another program.)

Sometimes, you might notice, you try to rerun a server and bind() fails, claiming "Address already in use." What does that mean? Well, a little bit of a socket that was connected is still hanging around in the kernel, and it's hogging the port. You can either wait for it to clear (a minute or so), or add code to your program allowing it to reuse the port, like this:

```
int yes=1;
//char yes='1'; // Solaris people use this

// lose the pesky "Address already in use" error message
if (setsockopt(listener,SOL_SOCKET,SO_REUSEADDR,&yes,sizeof(int)) == -1) {
    perror("setsockopt");
    exit(1);
}
```

One small extra final note about bind(): there are times when you won't absolutely have to call it. If you are connect()ing to a remote machine and you don't care what your local port is (as is the case with **telnet** where you only care about the remote port), you can simply call connect(), it'll check to see if the socket is unbound, and will bind() it to an unused local port if necessary.

4.3. connect()-Hey, you!

Let's just pretend for a few minutes that you're a telnet application. Your user commands you (just like in the movie *TRON*) to get a socket file descriptor. You comply and call socket(). Next, the user tells you to connect to "10.12.110.57" on port "23" (the standard telnet port.) Yow! What do you do now?

Lucky for you, program, you're now perusing the section on connect()—how to connect to a remote host. So read furiously onward! No time to lose!

The connect() call is as follows:

```
#include <sys/types.h>
#include <sys/socket.h>
int connect(int sockfd, struct sockaddr *serv_addr, int addrlen);
```

sockfd is our friendly neighborhood socket file descriptor, as returned by the socket() call, serv_addr is a struct sockaddr containing the destination port and IP address, and addrlen can be set to sizeof(struct sockaddr).

Isn't this starting to make more sense? Let's have an example:

```
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#define DEST_IP
                  "10.12.110.57"
#define DEST_PORT 23
main()
    int sockfd:
    struct sockaddr_in dest_addr; // will hold the destination addr
    sockfd = socket(PF_INET, SOCK_STREAM, 0); // do some error checking!
    dest_addr.sin_family = AF_INET;
                                             // host byte order
                                             // short, network byte order
    dest_addr.sin_port = htons(DEST_PORT);
    dest_addr.sin_addr.s_addr = inet_addr(DEST_IP);
    memset(&(dest_addr.sin_zero), '\0', 8); // zero the rest of the struct
```

```
// don't forget to error check the connect()!
connect(sockfd, (struct sockaddr *)&dest_addr, sizeof(struct sockaddr));
```

Again, be sure to check the return value from connect()-it'll return -1 on error and set the variable errno.

Also, notice that we didn't call bind(). Basically, we don't care about our local port number; we only care where we're going (the remote port). The kernel will choose a local port for us, and the site we connect to will automatically get this information from us. No worries.

4.4. listen()-Will somebody please call me?

Ok, time for a change of pace. What if you don't want to connect to a remote host. Say, just for kicks, that you want to wait for incoming connections and handle them in some way. The process is two step: first you listen(), then you accept() (see below.)

The listen call is fairly simple, but requires a bit of explanation:

```
int listen(int sockfd, int backlog);
```

sockfd is the usual socket file descriptor from the socket () system call. backlog is the number of connections allowed on the incoming queue. What does that mean? Well, incoming connections are going to wait in this queue until you accept() them (see below) and this is the limit on how many can queue up. Most systems silently limit this number to about 20; you can probably get away with setting it to 5 or 10.

Again, as per usual, listen() returns -1 and sets errno on error.

Well, as you can probably imagine, we need to call bind() before we call listen() or the kernel will have us listening on a random port. Bleah! So if you're going to be listening for incoming connections, the sequence of system calls you'll make is:

```
socket();
bind();
listen();
/* accept() goes here */
```

I'll just leave that in the place of sample code, since it's fairly self-explanatory. (The code in the accept () section, below, is more complete.) The really tricky part of this whole sha-bang is the call to accept().

4.5. accept()-"Thank you for calling port 3490."

Get ready-the accept() call is kinda weird! What's going to happen is this: someone far far away will try to connect() to your machine on a port that you are listen()ing on. Their connection will be queued up waiting to be accept()ed. You call accept() and you tell it to get the pending connection. It'll return to you a brand new socket file descriptor to use for this single connection! That's right, suddenly you have two socket file descriptors for the price of one! The original one is still listening on your port and the newly created one is finally ready to send() and recv(). We're there!

The call is as follows:

```
#include <sys/types.h>
#include <sys/socket.h>
```

```
int accept(int sockfd, struct sockaddr *addr, socklen_t *addrlen);
```

sockfd is the listen()ing socket descriptor. Easy enough. addr will usually be a pointer to a local struct sockaddr_in. This is where the information about the incoming connection will go (and with it you can determine which host is calling you from which port). addrlen is a local integer variable that should be set to sizeof(struct sockaddr_in) before its address is passed to accept(). Accept will not put more than that many bytes into addr. If it puts fewer in, it'll change the value of addrlen to reflect that.

Guess what? accept() returns -1 and sets errno if an error occurs. Betcha didn't figure that.

Like before, this is a bunch to absorb in one chunk, so here's a sample code fragment for your perusal:

```
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#define MYPORT 3490
                       // the port users will be connecting to
#define BACKLOG 10
                       // how many pending connections queue will hold
main()
    int sockfd, new_fd; // listen on sock_fd, new connection on new_fd
    struct sockaddr_in my_addr; // my address information
    struct sockaddr_in their_addr; // connector's address information
    int sin_size;
    sockfd = socket(PF_INET, SOCK_STREAM, 0); // do some error checking!
    mv_addr.sin_family = AF_INET;
                                          // host byte order
    my_addr.sin_port = htons(MYPORT);
                                         // short, network byte order
    my_addr.sin_addr.s_addr = INADDR_ANY; // auto-fill with my IP
   memset(&(my_addr.sin_zero), '\0', 8); // zero the rest of the struct
    // don't forget your error checking for these calls:
    bind(sockfd, (struct sockaddr *)&my_addr, sizeof(struct sockaddr));
    listen(sockfd, BACKLOG);
    sin_size = sizeof(struct sockaddr_in);
    new_fd = accept(sockfd, (struct sockaddr *)&their_addr, &sin_size);
```

Again, note that we will use the socket descriptor new_fd for all send() and recv() calls. If you're only getting one single connection ever, you can close() the listening sockfd in order to prevent more incoming connections on the same port, if you so desire.

4.6. send() and recv()-Talk to me, baby!

These two functions are for communicating over stream sockets or connected datagram sockets. If you want to use regular unconnected datagram sockets, you'll need to see the section on sendto() and recvfrom(), below.

The send() call:

```
int send(int sockfd, const void *msg, int len, int flags);
```

sockfd is the socket descriptor you want to send data to (whether it's the one returned by socket() or the one you got with accept().) msg is a pointer to the data you want to send, and len is the length of that data in bytes. Just set flags to 0. (See the send() man page for more information concerning flags.)

Some sample code might be:

```
char *msg = "Beej was here!";
int len, bytes_sent;
len = strlen(msg);
bytes_sent = send(sockfd, msg, len, 0);
```

send() returns the number of bytes actually sent out-this might be less than the number you told it to send! See, sometimes you tell it to send a whole gob of data and it just can't handle it. It'll fire off as much of the data as it can, and trust you to send the rest later. Remember, if the value returned by send() doesn't match the value in len, it's up to you to send the rest of the string. The good news is this: if the packet is small (less than 1K or so) it will *probably* manage to send the whole thing all in one go. Again, -1 is returned on error, and errno is set to the error number.

The recv() call is similar in many respects:

```
int recv(int sockfd, void *buf, int len, unsigned int flags);
```

sockfd is the socket descriptor to read from, buf is the buffer to read the information into, len is the maximum length of the buffer, and flags can again be set to 0. (See the recv() man page for flag information.)

recv() returns the number of bytes actually read into the buffer, or -1 on error (with errno set, accordingly.)

Wait! recv() can return 0. This can mean only one thing: the remote side has closed the connection on you! A return value of 0 is recv()'s way of letting you know this has occurred.

There, that was easy, wasn't it? You can now pass data back and forth on stream sockets! Whee! You're a Unix Network Programmer!

4.7. sendto() and recvfrom()-Talk to me, DGRAM-style

"This is all fine and dandy," I hear you saying, "but where does this leave me with unconnected datagram sockets?" No problemo, amigo. We have just the thing.

Since datagram sockets aren't connected to a remote host, guess which piece of information we need to give before we send a packet? That's right! The destination address! Here's the scoop:

```
int sendto(int sockfd, const void *msg, int len, unsigned int flags,
           const struct sockaddr *to, socklen_t tolen);
```

As you can see, this call is basically the same as the call to send() with the addition of two other pieces of information. to is a pointer to a struct sockaddr (which you'll probably have as a struct sockaddr_in and cast it at the last minute) which contains the destination IP address and port. tolen, an int deep-down, can simply be set to size of (struct sockaddr).

Just like with send(), sendto() returns the number of bytes actually sent (which, again, might be less than the number of bytes you told it to send!), or -1 on error.

Equally similar are recv() and recvfrom(). The synopsis of recvfrom() is:

```
int recvfrom(int sockfd, void *buf, int len, unsigned int flags,
             struct sockaddr *from, int *fromlen);
```

Again, this is just like recv() with the addition of a couple fields. from is a pointer to a local struct sockaddr that will be filled with the IP address and port of the originating machine. fromlen is a pointer to a local int that should be initialized to sizeof(struct sockaddr). When the function returns, from len will contain the length of the address actually stored in from.

recvfrom() returns the number of bytes received, or -1 on error (with errno set accordingly.)

Remember, if you connect() a datagram socket, you can then simply use send() and recv() for all your transactions. The socket itself is still a datagram socket and the packets still use UDP, but the socket interface will automatically add the destination and source information for you.

4.8. close() and shutdown()—Get outta my face!

Whew! You've been send()ing and recv()ing data all day long, and you've had it. You're ready to close the connection on your socket descriptor. This is easy. You can just use the regular Unix file descriptor close() function:

```
close(sockfd);
```

This will prevent any more reads and writes to the socket. Anyone attempting to read or write the socket on the remote end will receive an error.

Just in case you want a little more control over how the socket closes, you can use the shutdown() function. It allows you to cut off communication in a certain direction, or both ways (just like close() does.) Synopsis:

```
int shutdown(int sockfd, int how);
```

sockfd is the socket file descriptor you want to shutdown, and how is one of the following:

- 0 Further receives are disallowed
- 1 Further sends are disallowed
- 2 Further sends and receives are disallowed (like close())

shutdown() returns 0 on success, and -1 on error (with errno set accordingly.)

If you deign to use shutdown() on unconnected datagram sockets, it will simply make the socket unavailable for further send() and recv() calls (remember that you can use these if you connect() your datagram socket.)

It's important to note that shutdown() doesn't actually close the file descriptor-it just changes its usability. To free a socket descriptor, you need to use close().

Nothing to it.

4.9. getpeername()-Who are you?

This function is so easy.

It's so easy, I almost didn't give it it's own section. But here it is anyway.

The function getpeername() will tell you who is at the other end of a connected stream socket. The synopsis:

```
int getpeername(int sockfd, struct sockaddr *addr, int *addrlen);
```

sockfd is the descriptor of the connected stream socket, addr is a pointer to a struct sockaddr (or a struct sockaddr in) that will hold the information about the other side of the connection, and addrlen is a pointer to an int, that should be initialized to size of (struct sockaddr).

The function returns -1 on error and sets *errno* accordingly.

Once you have their address, you can use inet_ntoa() or gethostbyaddr() to print or get more information. No, you can't get their login name. (Ok, ok. If the other computer is running an ident daemon, this is possible. This, however, is beyond the scope of this document. Check out RFC-1413¹⁰ for more info.)

4.10. gethostname()-Who am I?

Even easier than getpeername() is the function gethostname(). It returns the name of the computer that your program is running on. The name can then be used by gethostbyname(), below, to determine the IP address of your local machine.

What could be more fun? I could think of a few things, but they don't pertain to socket programming. Anyway, here's the breakdown:

```
#include <unistd.h>
int gethostname(char *hostname, size_t size);
```

The arguments are simple: hostname is a pointer to an array of chars that will contain the hostname upon the function's return, and size is the length in bytes of the hostname array.

The function returns 0 on successful completion, and -1 on error, setting errno as usual.

4.11. DNS-You say "whitehouse.gov", I say "63.161.169.137"

In case you don't know what DNS is, it stands for "Domain Name Service". In a nutshell, you tell it what the human-readable address is for a site, and it'll give you the IP address (so you can use it with bind(), connect(), sendto(), or whatever you need it for.) This way, when someone enters:

```
$ telnet whitehouse.gov
```

telnet can find out that it needs to connect() to "63.161.169.137".

But how does it work? You'll be using the function gethostbyname():

```
#include <netdb.h>
struct hostent *gethostbyname(const char *name);
```

As you see, it returns a pointer to a struct hostent, the layout of which is as follows:

```
struct hostent {
    char
            *h_name;
    char
            **h_aliases;
    int
            h_addrtype;
            h_length;
    int
    char
            **h_addr_list;
#define h_addr h_addr_list[0]
```

And here are the descriptions of the fields in the struct hostent:

 $^{^{10}~{\}rm http://www.rfc\text{-}editor.org/rfc/rfc1413.txt}$

- h_name Official name of the host.
- h_aliases A NULL-terminated array of alternate names for the host.
- h_addrtype The type of address being returned; usually AF_INET.
- h_length The length of the address in bytes.
- h_addr_list A zero-terminated array of network addresses for the host. Host addresses are in Network Byte Order.
- h_addr The first address in h_addr_list.

gethostbyname() returns a pointer to the filled struct hostent, or NULL on error. (But errno is not set-h_errno is set instead. See herror(), below.)

But how is it used? Sometimes (as we find from reading computer manuals), just spewing the information at the reader is not enough. This function is certainly easier to use than it looks. Here's an example program¹¹:

```
** getip.c -- a hostname lookup demo
#include <stdio.h>
#include <stdlib.h>
#include <errno.h>
#include <netdb.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
int main(int argc, char *argv[])
    struct hostent *h;
    if (argc != 2) { // error check the command line
         fprintf(stderr, "usage: getip address\n");
    if ((h=gethostbyname(argv[1])) == NULL) { // get the host info
         herror("gethostbyname");
         exit(1);
    }
    \label{eq:printf("Host name : %s\n", h->h_name);}
    printf("IP Address : %s\n", inet_ntoa(*((struct in_addr *)h->h_addr)));
   return 0;
```

With gethostbyname(), you can't use perror() to print error message (since errno is not used). Instead, call herror().

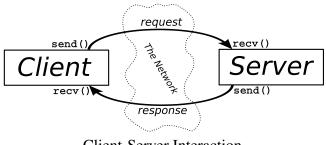
It's pretty straightforward. You simply pass the string that contains the machine name ("whitehouse.gov") to gethostbyname(), and then grab the information out of the returned struct hostent.

The only possible weirdness might be in the printing of the IP address, above. h->h_addr is a char*, but inet_ntoa() wants a struct in_addr passed to it. So I cast h->h_addr to a struct in_addr*, then dereference it to get at the data.

¹¹ http://beej.us/guide/bgnet/examples/getip.c

5. Client-Server Background

It's a client-server world, baby. Just about everything on the network deals with client processes talking to server processes and vice-versa. Take telnet, for instance. When you connect to a remote host on port 23 with telnet (the client), a program on that host (called telnetd, the server) springs to life. It handles the incoming telnet connection, sets you up with a login prompt, etc.



Client-Server Interaction.

The exchange of information between client and server is summarized in Figure 2.

Note that the client-server pair can speak SOCK_STREAM, SOCK_DGRAM, or anything else (as long as they're speaking the same thing.) Some good examples of client-server pairs are telnet/telnetd, ftp/ftpd, or bootp/bootpd. Every time you use ftp, there's a remote program, ftpd, that serves you.

Often, there will only be one server on a machine, and that server will handle multiple clients using fork(). The basic routine is: server will wait for a connection, accept() it, and fork() a child process to handle it. This is what our sample server does in the next section.

5.1. A Simple Stream Server

All this server does is send the string "Hello, World!\n" out over a stream connection. All you need to do to test this server is run it in one window, and telnet to it from another with:

```
$ telnet remotehostname 3490
```

where remotehostname is the name of the machine you're running it on.

The server code¹²: (Note: a trailing backslash on a line means that the line is continued on the next.)

```
** server.c -- a stream socket server demo
#include <stdio.h>
#include <stdlib.h>
#include <unistd.h>
#include <errno.h>
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#include <sys/wait.h>
#include <signal.h>
```

¹² http://beej.us/guide/bgnet/examples/server.c

```
#define MYPORT 3490
                        // the port users will be connecting to
#define BACKLOG 10
                        // how many pending connections queue will hold
void sigchld_handler(int s)
    while(waitpid(-1, NULL, WNOHANG) > 0);
}
int main(void)
    int sockfd, new_fd; // listen on sock_fd, new connection on new_fd
    struct sockaddr_in my_addr; // my address information
    struct sockaddr_in their_addr; // connector's address information
    socklen_t sin_size;
    struct sigaction sa;
    int yes=1;
    if ((sockfd = socket(PF_INET, SOCK_STREAM, 0)) == -1) {
        perror("socket");
        exit(1);
    }
    if (setsockopt(sockfd,SOL_SOCKET,SO_REUSEADDR,&yes,sizeof(int)) == -1) {
        perror("setsockopt");
        exit(1);
    }
    my_addr.sin_family = AF_INET;
                                           // host byte order
    my_addr.sin_port = htons(MYPORT);
                                           // short, network byte order
    my_addr.sin_addr.s_addr = INADDR_ANY; // automatically fill with my IP
    memset(&(my_addr.sin_zero), '\0', 8); // zero the rest of the struct
    if (bind(sockfd, (struct sockaddr *)&my_addr, sizeof(struct sockaddr))
                                                                      == -1) {
        perror("bind");
        exit(1);
    }
    if (listen(sockfd, BACKLOG) == -1) {
        perror("listen");
        exit(1);
    sa.sa_handler = sigchld_handler; // reap all dead processes
    sigemptyset(&sa.sa_mask);
    sa.sa_flags = SA_RESTART;
    if (sigaction(SIGCHLD, &sa, NULL) == -1) {
        perror("sigaction");
        exit(1);
    }
    while(1) { // main accept() loop
        sin_size = sizeof(struct sockaddr_in);
        if ((new_fd = accept(sockfd, (struct sockaddr *)&their_addr,
                                                         &sin_size)) == -1) {
            perror("accept");
            continue;
        }
        printf("server: got connection from %s\n",
                                            inet_ntoa(their_addr.sin_addr));
        if (!fork()) { // this is the child process
            close(sockfd); // child doesn't need the listener
if (send(new_fd, "Hello, world!\n", 14, 0) == -1)
                perror("send");
            close(new_fd);
            exit(0);
        close(new_fd); // parent doesn't need this
    }
    return 0;
```

}

In case you're curious, I have the code in one big main() function for (I feel) syntactic clarity. Feel free to split it into smaller functions if it makes you feel better.

(Also, this whole sigaction() thing might be new to you—that's ok. The code that's there is responsible for reaping zombie processes that appear as the fork()ed child processes exit. If you make lots of zombies and don't reap them, your system administrator will become agitated.)

You can get the data from this server by using the client listed in the next section.

5.2. A Simple Stream Client

This guy's even easier than the server. All this client does is connect to the host you specify on the command line, port 3490. It gets the string that the server sends.

The client source¹³:

```
** client.c -- a stream socket client demo
#include <stdio.h>
#include <stdlib.h>
#include <unistd.h>
#include <errno.h>
#include <string.h>
#include <netdb.h>
#include <sys/types.h>
#include <netinet/in.h>
#include <sys/socket.h>
#define PORT 3490 // the port client will be connecting to
#define MAXDATASIZE 100 // max number of bytes we can get at once
int main(int argc, char *argv[])
{
    int sockfd, numbytes;
    char buf[MAXDATASIZE]:
    struct hostent *he;
    struct sockaddr_in their_addr; // connector's address information
    if (argc != 2) {
        fprintf(stderr, "usage: client hostname\n");
        exit(1);
    }
    if ((he=gethostbyname(argv[1])) == NULL) { // get the host info
        herror("gethostbyname");
        exit(1);
    if ((sockfd = socket(PF_INET, SOCK_STREAM, 0)) == -1) {
        perror("socket");
        exit(1);
    their_addr.sin_family = AF_INET;
                                          // host byte order
    their_addr.sin_port = htons(PORT); // short, network byte order
    their_addr.sin_addr = *((struct in_addr *)he->h_addr);
    memset(&(their_addr.sin_zero), '\0', 8); // zero the rest of the struct
    if (connect(sockfd, (struct sockaddr *)&their_addr,
                                             sizeof(struct sockaddr)) == -1) {
        perror("connect");
        exit(1);
    }
```

 $^{^{13}}$ http://beej.us/guide/bgnet/examples/client.c

```
if ((numbytes=recv(sockfd, buf, MAXDATASIZE-1, 0)) == -1) {
        perror("recv");
        exit(1);
    buf[numbytes] = '\0';
    printf("Received: %s",buf);
    close(sockfd);
    return 0;
}
```

Notice that if you don't run the server before you run the client, connect() returns "Connection refused". Very useful.

5.3. Datagram Sockets

I really don't have that much to talk about here, so I'll just present a couple of sample programs: talker.c and listener.c.

listener sits on a machine waiting for an incoming packet on port 4950. **talker** sends a packet to that port, on the specified machine, that contains whatever the user enters on the command line.

Here is the source for **listener.c**¹⁴:

```
** listener.c -- a datagram sockets "server" demo
#include <stdio.h>
#include <stdlib.h>
#include <unistd.h>
#include <errno.h>
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#define MYPORT 4950
                          // the port users will be connecting to
#define MAXBUFLEN 100
int main(void)
    int sockfd:
    struct sockaddr_in my_addr;
                                        // my address information
    struct sockaddr_in their_addr; // connector's address information
     socklen_t addr_len;
    int numbytes;
    char buf [MAXBUFLEN];
     if ((sockfd = socket(PF_INET, SOCK_DGRAM, 0)) == -1) {
         perror("socket");
         exit(1);
    my_addr.sin_family = AF_INET;
                                                 // host byte order
    my_addr.sin_port = htons(MYPORT);  // short, network byte order
my_addr.sin_addr.s_addr = INADDR_ANY; // automatically fill with my IP
    memset(&(my_addr.sin_zero), '\0', 8); // zero the rest of the struct
     if (bind(sockfd, (struct sockaddr *)&my_addr,
         sizeof(struct sockaddr)) == -1) {
         perror("bind");
         exit(1);
    }
```

¹⁴ http://beej.us/guide/bgnet/examples/listener.c

```
addr_len = sizeof(struct sockaddr);
if ((numbytes=recvfrom(sockfd, buf, MAXBUFLEN-1 , 0,
         (struct sockaddr *)&their_addr, &addr_len)) == -1) {
         perror("recvfrom");
         exit(1);
    printf("got packet from %s\n",inet_ntoa(their_addr.sin_addr));
    printf("packet is %d bytes long\n",numbytes);
    buf[numbytes] = '\0';
    printf("packet contains \"%s\"\n",buf);
    close(sockfd);
    return 0;
}
```

Notice that in our call to socket() we're finally using SOCK_DGRAM. Also, note that there's no need to listen() or accept(). This is one of the perks of using unconnected datagram sockets!

Next comes the source for **talker.c**¹⁵:

```
** talker.c -- a datagram "client" demo
#include <stdio.h>
#include <stdlib.h>
#include <unistd.h>
#include <errno.h>
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#include <netdb.h>
#define SERVERPORT 4950
                             // the port users will be connecting to
int main(int argc, char *argv[])
{
    int sockfd:
    struct sockaddr_in their_addr; // connector's address information
    struct hostent *he;
    int numbytes;
    if (argc != 3) {
        fprintf(stderr, "usage: talker hostname message\n");
        exit(1);
    }
    if ((he=gethostbyname(argv[1])) == NULL) { // get the host info
        perror("gethostbyname");
        exit(1);
    if ((sockfd = socket(AF_INET, SOCK_DGRAM, 0)) == -1) {
        perror("socket");
        exit(1);
    }
    their_addr.sin_family = AF_INET;
                                             // host byte order
    their_addr.sin_port = htons(SERVERPORT); // short, network byte order
    their_addr.sin_addr = *((struct in_addr *)he->h_addr);
    memset(&(their_addr.sin_zero), '\0', 8); // zero the rest of the struct
    if ((numbytes = sendto(sockfd, argv[2], strlen(argv[2]), 0,
              (struct sockaddr *)&their_addr, sizeof(struct sockaddr))) == -1) {
        perror("sendto");
```

 $^{^{15} \ \}mathtt{http://beej.us/guide/bgnet/examples/talker.c}$

```
exit(1);
    printf("sent %d bytes to %s\n", numbytes, inet_ntoa(their_addr.sin_addr));
    close(sockfd);
    return 0;
}
```

And that's all there is to it! Run listener on some machine, then run talker on another. Watch them communicate! Fun G-rated excitement for the entire nuclear family!

Except for one more tiny detail that I've mentioned many times in the past: connected datagram sockets. I need to talk about this here, since we're in the datagram section of the document. Let's say that talker calls connect() and specifies the listener's address. From that point on, talker may only sent to and receive from the address specified by connect(). For this reason, you don't have to use sendto() and recvfrom(); you can simply use send() and recv().

6. Slightly Advanced Techniques

These aren't *really* advanced, but they're getting out of the more basic levels we've already covered. In fact, if you've gotten this far, you should consider yourself fairly accomplished in the basics of Unix network programming! Congratulations!

So here we go into the brave new world of some of the more esoteric things you might want to learn about sockets. Have at it!

6.1. Blocking

Blocking. You've heard about it-now what the heck is it? In a nutshell, "block" is techie jargon for "sleep". You probably noticed that when you run listener, above, it just sits there until a packet arrives. What happened is that it called recvfrom(), there was no data, and so recvfrom() is said to "block" (that is, sleep there) until some data arrives.

Lots of functions block. accept() blocks. All the recv() functions block. The reason they can do this is because they're allowed to. When you first create the socket descriptor with socket(), the kernel sets it to blocking. If you don't want a socket to be blocking, you have to make a call to fcntl():

```
#include <unistd.h>
#include <fcntl.h>
sockfd = socket(PF_INET, SOCK_STREAM, 0);
fcntl(sockfd, F_SETFL, O_NONBLOCK);
```

By setting a socket to non-blocking, you can effectively "poll" the socket for information. If you try to read from a non-blocking socket and there's no data there, it's not allowed to block-it will return -1 and errno will be set to EWOULDBLOCK.

Generally speaking, however, this type of polling is a bad idea. If you put your program in a busy-wait looking for data on the socket, you'll suck up CPU time like it was going out of style. A more elegant solution for checking to see if there's data waiting to be read comes in the following section on select().

6.2. select()-Synchronous I/O Multiplexing

This function is somewhat strange, but it's very useful. Take the following situation: you are a server and you want to listen for incoming connections as well as keep reading from the connections you already have.

No problem, you say, just an accept() and a couple of recv()s. Not so fast, buster! What if you're blocking on an accept() call? How are you going to recv() data at the same time? "Use non-blocking sockets!" No way! You don't want to be a CPU hog. What, then?

select() gives you the power to monitor several sockets at the same time. It'll tell you which ones are ready for reading, which are ready for writing, and which sockets have raised exceptions, if you really want to know that.

Without any further ado, I'll offer the synopsis of select():

```
#include <sys/time.h>
#include <sys/types.h>
#include <unistd.h>
```

```
int select(int numfds, fd_set *readfds, fd_set *writefds,
           fd_set *exceptfds, struct timeval *timeout);
```

The function monitors "sets" of file descriptors; in particular readfds, writefds, and exceptfds. If you want to see if you can read from standard input and some socket descriptor, sockfd, just add the file descriptors 0 and sockfd to the set readfds. The parameter numfds should be set to the values of the highest file descriptor plus one. In this example, it should be set to sockfd+1, since it is assuredly higher than standard input (0).

When select() returns, readfds will be modified to reflect which of the file descriptors you selected which is ready for reading. You can test them with the macro FD_ISSET(), below.

Before progressing much further, I'll talk about how to manipulate these sets. Each set is of the type fd_set. The following macros operate on this type:

- FD_ZERO(fd_set *set) clears a file descriptor set
- FD_SET(int fd, fd_set *set) adds fd to the set
- FD_CLR(int fd, fd_set *set) removes fd from the set
- FD_ISSET(int fd, fd_set *set) tests to see if fd is in the set

Finally, what is this weirded out struct timeval? Well, sometimes you don't want to wait forever for someone to send you some data. Maybe every 96 seconds you want to print "Still Going..." to the terminal even though nothing has happened. This time structure allows you to specify a timeout period. If the time is exceeded and select() still hasn't found any ready file descriptors, it'll return so you can continue processing.

The struct timeval has the follow fields:

```
struct timeval {
                    // seconds
    int tv_sec;
                    // microseconds
};
```

Just set tv_sec to the number of seconds to wait, and set tv_usec to the number of microseconds to wait. Yes, that's microseconds, not milliseconds. There are 1,000 microseconds in a millisecond, and 1,000 milliseconds in a second. Thus, there are 1,000,000 microseconds in a second. Why is it "usec"? The "u" is supposed to look like the Greek letter μ (Mu) that we use for "micro". Also, when the function returns, timeout might be updated to show the time still remaining. This depends on what flavor of Unix you're running.

Yay! We have a microsecond resolution timer! Well, don't count on it. Standard Unix timeslice is around 100 milliseconds, so you might have to wait that long no matter how small you set your struct timeval.

Other things of interest: If you set the fields in your struct timeval to 0, select() will timeout immediately, effectively polling all the file descriptors in your sets. If you set the parameter timeout to NULL, it will never timeout, and will wait until the first file descriptor is ready. Finally, if you don't care about waiting for a certain set, you can just set it to NULL in the call to select().

The following code snippet¹⁶ waits 2.5 seconds for something to appear on standard input:

```
** select.c -- a select() demo
#include <stdio.h>
#include <sys/time.h>
#include <sys/types.h>
#include <unistd.h>
```

 $^{^{16}}$ http://beej.us/guide/bgnet/examples/select.c

```
#define STDIN 0 // file descriptor for standard input
int main(void)
   struct timeval tv;
   fd_set readfds;
    tv.tv sec = 2:
   tv.tv_usec = 500000:
   FD_ZERO(&readfds);
   FD_SET(STDIN, &readfds);
    // don't care about writefds and exceptfds:
   select(STDIN+1, &readfds, NULL, NULL, &tv);
    if (FD_ISSET(STDIN, &readfds))
       printf("A key was pressed!\n");
       printf("Timed out.\n");
   return 0;
}
```

If you're on a line buffered terminal, the key you hit should be RETURN or it will time out anyway.

Now, some of you might think this is a great way to wait for data on a datagram socket-and you are right: it might be. Some Unices can use select in this manner, and some can't. You should see what your local man page says on the matter if you want to attempt it.

Some Unices update the time in your struct timeval to reflect the amount of time still remaining before a timeout. But others do not. Don't rely on that occurring if you want to be portable. (Use gettimeofday() if you need to track time elapsed. It's a bummer, I know, but that's the way it is.)

What happens if a socket in the read set closes the connection? Well, in that case, select() returns with that socket descriptor set as "ready to read". When you actually do recv() from it, recv() will return 0. That's how you know the client has closed the connection.

One more note of interest about select(): if you have a socket that is listen()ing, you can check to see if there is a new connection by putting that socket's file descriptor in the readfds set.

And that, my friends, is a quick overview of the almighty select() function.

But, by popular demand, here is an in-depth example. Unfortunately, the difference between the dirt-simple example, above, and this one here is significant. But have a look, then read the description that follows it.

This program¹⁷ acts like a simple multi-user chat server. Start it running in one window, then telnet to it ("telnet hostname 9034") from multiple other windows. When you type something in one **telnet** session, it should appear in all the others.

```
** selectserver.c -- a cheezy multiperson chat server
#include <stdio.h>
#include <stdlib.h>
#include <string.h>
#include <unistd.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#define PORT 9034 // port we're listening on
```

 $^{^{17}}$ http://beej.us/guide/bgnet/examples/selectserver.c

```
int main(void)
{
    fd_set master;
                    // master file descriptor list
    fd_set read_fds; // temp file descriptor list for select()
    struct sockaddr_in myaddr;  // server address
struct sockaddr_in remoteaddr; // client address
                      // maximum file descriptor number
    int fdmax;
                      // listening socket descriptor
    int listener;
                      // newly accept()ed socket descriptor
    int newfd:
    char buf [256];
                      // buffer for client data
    int nbytes;
    int yes=1;
                      // for setsockopt() SO_REUSEADDR, below
    socklen_t addrlen;
    int i, j;
    FD_ZERO(&master);
                          // clear the master and temp sets
    FD_ZERO(&read_fds);
    // get the listener
    if ((listener = socket(PF_INET, SOCK_STREAM, 0)) == -1) {
        perror("socket");
        exit(1);
    }
    // lose the pesky "address already in use" error message
    if (setsockopt(listener, SOL_SOCKET, SO_REUSEADDR, &yes,
                                                           sizeof(int)) == -1) {
        perror("setsockopt");
        exit(1);
    // bind
    myaddr.sin_family = AF_INET;
    myaddr.sin_addr.s_addr = INADDR_ANY;
    myaddr.sin_port = htons(PORT);
    memset(&(myaddr.sin_zero), '\0', 8);
    if (bind(listener, (struct sockaddr *)&myaddr, sizeof(myaddr)) == -1) {
        perror("bind");
        exit(1);
    }
    // listen
    if (listen(listener, 10) == -1) {
        perror("listen");
        exit(1);
    }
    // add the listener to the master set
    FD_SET(listener, &master);
    // keep track of the biggest file descriptor
    fdmax = listener; // so far, it's this one
    // main loop
    for(;;) {
        read_fds = master; // copy it
        if (select(fdmax+1, &read_fds, NULL, NULL, NULL) == -1) {
            perror("select");
            exit(1);
        // run through the existing connections looking for data to read
        for(i = 0; i <= fdmax; i++) {</pre>
            if (FD_ISSET(i, &read_fds)) { // we got one!!
                if (i == listener) {
                     // handle new connections
                     addrlen = sizeof(remoteaddr);
                     if ((newfd = accept(listener, (struct sockaddr *)&remoteaddr,
                                                                &addrlen)) == -1) {
                         perror("accept");
                     } else {
```

```
FD_SET(newfd, &master); // add to master set
                        if (newfd > fdmax) {
                                                 // keep track of the maximum
                            fdmax = newfd;
                        printf("selectserver: new connection from %s on "
                             "socket %d\n", inet_ntoa(remoteaddr.sin_addr), newfd);
                    }
                } else {
                    // handle data from a client
                    if ((nbytes = recv(i, buf, sizeof(buf), 0)) <= 0) {</pre>
                        // got error or connection closed by client
                        if (nbytes == 0) {
                            // connection closed
                            printf("selectserver: socket %d hung up\n", i);
                        } else {
                            perror("recv");
                        close(i); // bye!
                        FD_CLR(i, &master); // remove from master set
                        // we got some data from a client
                        for(j = 0; j \le fdmax; j++) {
                            // send to everyone!
                            if (FD_ISSET(j, &master)) {
                                 // except the listener and ourselves
                                 if (j != listener && j != i) {
                                     if (send(j, buf, nbytes, 0) == -1) {
                                         perror("send");
                                     }
                            }
                        }
                } // it's SO UGLY!
        }
    return 0;
}
```

Notice I have two file descriptor sets in the code: master and read_fds. The first, master, holds all the socket descriptors that are currently connected, as well as the socket descriptor that is listening for new connections.

The reason I have the master set is that select() actually changes the set you pass into it to reflect which sockets are ready to read. Since I have to keep track of the connections from one call of select() to the next, I must store these safely away somewhere. At the last minute, I copy the master into the read_fds, and then call select().

But doesn't this mean that every time I get a new connection, I have to add it to the master set? Yup! And every time a connection closes, I have to remove it from the master set? Yes, it does.

Notice I check to see when the listener socket is ready to read. When it is, it means I have a new connection pending, and I accept() it and add it to the master set. Similarly, when a client connection is ready to read, and recv() returns 0, I know the client has closed the connection, and I must remove it from the master set.

If the client recv() returns non-zero, though, I know some data has been received. So I get it, and then go through the master list and send that data to all the rest of the connected clients. And that, my friends, is a less-than-simple overview of the almighty select() function.

6.3. Handling Partial send()s

Remember back in the section about send(), above, when I said that send() might not send all the bytes you asked it to? That is, you want it to send 512 bytes, but it returns 412. What happened to the remaining 100 bytes?

Well, they're still in your little buffer waiting to be sent out. Due to circumstances beyond your control, the kernel decided not to send all the data out in one chunk, and now, my friend, it's up to you to get the data out there.

You could write a function like this to do it, too:

```
#include <sys/types.h>
#include <sys/socket.h>
int sendall(int s, char *buf, int *len)
    int total = 0;
                       // how many bytes we've sent
    int bytesleft = *len; // how many we have left to send
    int n;
    while(total < *len) {
        n = send(s, buf+total, bytesleft, 0);
        if (n == -1) { break; }
        total += n;
bytesleft -= n;
    *len = total; // return number actually sent here
    return n==-1?-1:0; // return -1 on failure, 0 on success
}
```

In this example, s is the socket you want to send the data to, buf is the buffer containing the data, and len is a pointer to an int containing the number of bytes in the buffer.

The function returns -1 on error (and errno is still set from the call to send().) Also, the number of bytes actually sent is returned in len. This will be the same number of bytes you asked it to send, unless there was an error. sendall() will do it's best, huffing and puffing, to send the data out, but if there's an error, it gets back to you right away.

For completeness, here's a sample call to the function:

```
char buf[10] = "Beej!";
int len;
len = strlen(buf);
if (sendall(s, buf, &len) == -1) {
   perror("sendall");
   printf("We only sent %d bytes because of the error!\n", len);
}
```

What happens on the receiver's end when part of a packet arrives? If the packets are variable length, how does the receiver know when one packet ends and another begins? Yes, real-world scenarios are a royal pain in the donkeys. You probably have to encapsulate (remember that from the data encapsulation section way back there at the beginning?) Read on for details!

6.4. Son of Data Encapsulation

What does it really mean to encapsulate data, anyway? In the simplest case, it means you'll stick a header on there with either some identifying information or a packet length, or both.

What should your header look like? Well, it's just some binary data that represents whatever you feel is necessary to complete your project.

Wow. That's vague.

Okay. For instance, let's say you have a multi-user chat program that uses SOCK_STREAMs. When a user types ("says") something, two pieces of information need to be transmitted to the server: what was said and who said it.

So far so good? "What's the problem?" you're asking.

The problem is that the messages can be of varying lengths. One person named "tom" might say, "Hi", and another person named "Benjamin" might say, "Hey guys what is up?"

So you send() all this stuff to the clients as it comes in. Your outgoing data stream looks like this:

```
tom HiBenjamin Heyguyswhatisup?
```

And so on. How does the client know when one message starts and another stops? You could, if you wanted, make all messages the same length and just call the sendall() we implemented, above. But that wastes bandwidth! We don't want to send() 1024 bytes just so "tom" can say "Hi".

So we *encapsulate* the data in a tiny header and packet structure. Both the client and server know how to pack and unpack (sometimes referred to as "marshal" and "unmarshal") this data. Don't look now, but we're starting to define a protocol that describes how a client and server communicate!

In this case, let's assume the user name is a fixed length of 8 characters, padded with '\0'. And then let's assume the data is variable length, up to a maximum of 128 characters. Let's have a look a sample packet structure that we might use in this situation:

- 1. len (1 byte, unsigned) The total length of the packet, counting the 8-byte user name and chat data.
- 2. name (8 bytes) The user's name, NUL-padded if necessary.
- 3. chatdata (n-bytes) The data itself, no more than 128 bytes. The length of the packet should be calculated as the length of this data plus 8 (the length of the name field, above).

Why did I choose the 8-byte and 128-byte limits for the fields? I pulled them out of the air, assuming they'd be long enough. Maybe, though, 8 bytes is too restrictive for your needs, and you can have a 30-byte name field, or whatever. The choice is up to you.

Using the above packet definition, the first packet would consist of the following information (in hex and ASCII):

```
74 6F 6D 00 00 00 00 00 T o m (padding)
   OΑ
 (length)
And the second is similar:
           42 65 6E 6A 61 6D 69 6E
                                            48 65 79 20 67 75 79 73 20 77 ...
 (length) B e n
```

(The length is stored in Network Byte Order, of course. In this case, it's only one byte so it doesn't matter, but generally speaking you'll want all your binary integers to be stored in Network Byte Order in your packets.)

When you're sending this data, you should be safe and use a command similar to sendall(), above, so you know all the data is sent, even if it takes multiple calls to send() to get it all out.

Likewise, when you're receiving this data, you need to do a bit of extra work. To be safe, you should assume that you might receive a partial packet (like maybe we receive "14 42 65 6E" from Benjamin, above, but that's all we get in this call to recv()). We need to call recv() over and over again until the packet is completely received.

But how? Well, we know the number of bytes we need to receive in total for the packet to be complete, since that number is tacked on the front of the packet. We also know the maximum packet size is 1+8+128, or 137 bytes (because that's how we defined the packet.)

What you can do is declare an array big enough for two packets. This is your work array where you will reconstruct packets as they arrive.

Every time you recv() data, you'll feed it into the work buffer and check to see if the packet is complete. That is, the number of bytes in the buffer is greater than or equal to the length specified in the header (+1, because the length in the header doesn't include the byte for the length itself.) If the number of bytes in the buffer is less than 1, the packet is not complete, obviously. You have to make a special case for this, though, since the first byte is garbage and you can't rely on it for the correct packet length.

Once the packet is complete, you can do with it what you will. Use it, and remove it from your work buffer.

Whew! Are you juggling that in your head yet? Well, here's the second of the one-two punch: you might have read past the end of one packet and onto the next in a single recv() call. That is, you have a work buffer with one complete packet, and an incomplete part of the next packet! Bloody heck. (But this is why you made your work buffer large enough to hold two packets—in case this happened!)

Since you know the length of the first packet from the header, and you've been keeping track of the number of bytes in the work buffer, you can subtract and calculate how many of the bytes in the work buffer belong to the second (incomplete) packet. When you've handled the first one, you can clear it out of the work buffer and move the partial second packed down the to front of the buffer so it's all ready to go for the next recv().

(Some of you readers will note that actually moving the partial second packet to the beginning of the work buffer takes time, and the program can be coded to not require this by using a circular buffer. Unfortunately for the rest of you, a discussion on circular buffers is beyond the scope of this article. If you're still curious, grab a data structures book and go from there.)

I never said it was easy. Ok, I did say it was easy. And it is; you just need practice and pretty soon it'll come to you naturally. By Excalibur I swear it!

6.5. Broadcast Packets-Hello, World!

So far, this guide has talked about sending data from one host to one other host. But it is possible, I insist, that you can, with the proper authority, send data to multiple hosts at the same time!

With UDP (only UDP, not TCP) and standard IPv4, this is done through a mechanism called broadcasting. With IPv6 (not appearing in this guide...yet), broadcasting isn't supported, and you have to resort to the often superior technique of *multicasting*. But enough of the starry-eyed future—we're stuck in the 32-bit present.

But wait! You can't just run off and start broadcasting willy-nilly; You have to set the socket option SO_BROADCAST before you can send a broadcast packet out on the network. It's like a one of those little plastic covers they put over the missile launch switch! That's just how much power you hold in your hands!

But seriously, though, there is a danger to using broadcast packets, and that is: every system that recieves a broadcast packet must undo all the onion-skin layers of data encapsulation until it finds out what port the data is destined to. And then it hands the data over or discards it. In either case, it's a lot of work for each machine that recieves the broadcast packet, and since it is all of them on the local network, that could be a lot of machines doing a lot of unnecessary work. When the game Doom first came out, this was a complaint about its network code.

Yes, I said the local network. There is more than one way to skin a cat... wait a minute. Is there really more than one way to skin a cat? What kind of expression is that? Uh, and likewise, there is more than one way to send a broadcast packet, but the broadcast packets will usually be restricted to your local network no matter how you send them.

So now to the meat and potatoes of the whole thing: how do you specify the destination address for a broadcast message? There are two common ways.

- 1. Send the data to your broadcast address. This is your network number with all one-bits set for the host portion of the address. For instance, at home my network is 192.168.1.0, my netmask is 255.255.255.0, so the last byte of the address is my host number (because the first three bytes, according to the netmask, are the network number). So my broadcast address is 192.168.1.255. Under Unix, the **ifconfig** command will actually give you all this data. (If you're curious, the bitwise logic to get your broadcast address is network_number OR (NOT netmask).)
- 2. Send the data to the "global" broadcast address. This is 255.255.255.255, aka IN-ADDR_BROADCAST. Many machines will automatically bitwise AND this with your network number to convert it to a network broadcast address, but some won't. It varies.

So what happens if you try to send data on the broadcast address without first setting the SO_BROADCAST socket option? Well, let's fire up good old talker and listener and see what happens.

```
$ talker 192.168.1.2 foo
sent 3 bytes to 192.168.1.2
$ talker 192.168.1.255 foo
sendto: Permission denied
$ talker 255.255.255.255 foo
sendto: Permission denied
```

Yes, it's not happy at all...because we didn't set the SO_BROADCAST socket option. Do that, and now you can sendto() anywhere you want!

In fact, that's the *only difference* between a UDP application that can broadcast and one that can't. So let's take the old talker application and add one section that sets the SO_BROADCAST socket option. We'll call this program **broadcaster.c**¹⁸:

```
** broadcaster.c -- a datagram "client" like talker.c, except
                      this one can broadcast
#include <stdio.h>
#include <stdlib.h>
#include <unistd.h>
#include <urno.h>
#include <string.h>
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#include <netdb.h>
#define SERVERPORT 4950
                              // the port users will be connecting to
int main(int argc, char *argv[])
{
    int sockfd;
    struct sockaddr_in their_addr; // connector's address information
    struct hostent *he;
    int numbytes;
    int broadcast = 1;
    //char broadcast = '1'; // if that doesn't work, try this
        fprintf(stderr, "usage: broadcaster hostname message\n");
        exit(1):
    }
    if ((he=gethostbyname(argv[1])) == NULL) { // get the host info
        perror("gethostbyname");
        exit(1);
    }
```

 $^{^{18} \ \}mathtt{http://beej.us/guide/bgnet/examples/broadcaster.c}$

```
if ((sockfd = socket(AF_INET, SOCK_DGRAM, 0)) == -1) {
        perror("socket");
        exit(1);
    }
    // this call is the difference between this program and talker.c:
    if (setsockopt(sockfd, SOL_SOCKET, SO_BROADCAST, &broadcast,
        sizeof(broadcast)) == -1) {
        perror("setsockopt (SO_BROADCAST)");
        exit(1);
    their_addr.sin_family = AF_INET;
                                         // host byte order
    their_addr.sin_port = htons(SERVERPORT); // short, network byte order
    their_addr.sin_addr = *((struct in_addr *)he->h_addr);
   memset(&(their_addr.sin_zero), '\0', 8); // zero the rest of the struct
    if ((numbytes=sendto(sockfd, argv[2], strlen(argv[2]), 0,
             (struct sockaddr *)&their_addr, sizeof(struct sockaddr))) == -1) {
        perror("sendto");
        exit(1);
   printf("sent %d bytes to %s\n", numbytes, inet_ntoa(their_addr.sin_addr));
    close(sockfd);
   return 0;
}
```

What's different between this and a "normal" UDP client/server situation? Nothing! (With the exception of the client being allowed to send broadcast packets in this case.) As such, go ahead and run the old UDP listener program in one window, and broadcaster in another. You should be now be able to do all those sends that failed, above.

```
$ talker 192.168.1.2 foo
sent 3 bytes to 192.168.1.2
$ talker 192.168.1.255 foo
sent 3 bytes to 192.168.1.255
$ talker 255.255.255.255 foo
sent 3 bytes to 255.255.255.255
```

And you should see **listener** responding that it got the packets.

Well, that's kind of exciting. But now fire up **listener** on another machine next to you on the same network so that you have two copies going, one on each machine, and run broadcaster again with your broadcast address... Hey! Both listeners get the packet even though you only called sendto() once! Cool!

If the **listener** gets data you send directly to it, but not data on the broadcast address, it could be that you have a firewall on your local machine that is blocking the packets. (Yes, Pat and Bapper, thank you for realizing before I did that this is why my sample code wasn't working. I told you I'd mention you in the guide, and here you are. So *nyah*.)

Again, be careful with broadcast packets. Since every machine on the LAN will be forced to deal with the packet whether it recvfrom()s it or not, it can present quite a load to the entire computing network. They are definitely to be used sparingly and appropriately.

7. Common Questions

Where can I get those header files?

If you don't have them on your system already, you probably don't need them. Check the manual for your particular platform. If you're building for Windows, you only need to #include <winsock.h>.

What do I do when bind() reports "Address already in use"?

You have to use setsockopt() with the SO_REUSEADDR option on the listening socket. Check out the section on bind() and the section on select() for an example.

How do I get a list of open sockets on the system?

Use the **netstat**. Check the **man** page for full details, but you should get some good output just typing:

\$ netstat

The only trick is determining which socket is associated with which program. :-)

How can I view the routing table?

Run the **route** command (in /sbin on most Linuxes) or the command **netstat -r**.

How can I run the client and server programs if I only have one computer? Don't I need a network to write network program?

Fortunately for you, virtually all machines implement a loopback network "device" that sits in the kernel and pretends to be a network card. (This is the interface listed as "lo" in the routing table.)

Pretend you're logged into a machine named "goat". Run the client in one window and the server in another. Or start the server in the background ("server &") and run the client in the same window. The upshot of the loopback device is that you can either client goat or client localhost (since "localhost" is likely defined in your /etc/hosts file) and you'll have the client talking to the server without a network!

In short, no changes are necessary to any of the code to make it run on a single non-networked machine! Huzzah!

How can I tell if the remote side has closed connection?

You can tell because recv() will return 0.

How do I implement a "ping" utility? What is ICMP? Where can I find out more about raw sockets and SOCK_RAW?

All your raw sockets questions will be answered in W. Richard Stevens' UNIX Network Programming books. See the books section of this guide.

How do I build for Windows?

First, delete Windows and install Linux or BSD. };-). No, actually, just see the section on building for Windows in the introduction.

How do I build for Solaris/SunOS? I keep getting linker errors when I try to compile!

The linker errors happen because Sun boxes don't automatically compile in the socket libraries. See the section on building for Solaris/SunOS in the introduction for an example of how to do this.

Why does select() keep falling out on a signal?

Signals tend to cause blocked system calls to return -1 with errno set to EINTR. When you set up a signal handler with sigaction(), you can set the flag SA_RESTART, which is supposed to restart the system call after it was interrupted.

Naturally, this doesn't always work.

My favorite solution to this involves a goto statement. You know this irritates your professors to no end, so go for it!

```
select_restart:
if ((err = select(fdmax+1, &readfds, NULL, NULL, NULL)) == -1) {
    if (errno == EINTR) {
        // some signal just interrupted us, so restart
        goto select_restart;
    // handle the real error here:
   perror("select");
}
```

Sure, you don't *need* to use goto in this case; you can use other structures to control it. But I think the goto statement is actually cleaner.

How can I implement a timeout on a call to recv()?

Use select()! It allows you to specify a timeout parameter for socket descriptors that you're looking to read from. Or, you could wrap the entire functionality in a single function, like this:

```
#include <unistd.h>
#include <sys/time.h>
#include <sys/types.h>
#include <sys/socket.h>
int recvtimeout(int s, char *buf, int len, int timeout)
    fd_set fds;
    int n;
    struct timeval tv;
    // set up the file descriptor set
    FD_ZERO(&fds);
    FD_SET(s, &fds);
    // set up the struct timeval for the timeout
    tv.tv_sec = timeout;
    tv.tv_usec = 0;
    // wait until timeout or data received
    n = select(s+1, &fds, NULL, NULL, &tv);
    if (n == 0) return -2; // timeout!
    if (n == -1) return -1; // error
    // data must be here, so do a normal recv()
    return recv(s, buf, len, 0);
}
// Sample call to recvtimeout():
n = recvtimeout(s, buf, sizeof(buf), 10); // 10 second timeout
if (n == -1) {
    // error occurred
    perror("recvtimeout");
else if (n == -2) {
// timeout occurred
} else {
    // got some data in buf
}
```

Notice that recvtimeout() returns -2 in case of a timeout. Why not return 0? Well, if you recall, a return value of 0 on a call to recv() means that the remote side closed the connection. So that return value is already spoken for, and -1 means "error", so I chose -2 as my timeout indicator.

How do I encrypt or compress the data before sending it through the socket?

One easy way to do encryption is to use SSL (secure sockets layer), but that's beyond the scope of this guide.

But assuming you want to plug in or implement your own compressor or encryption system, it's just a matter of thinking of your data as running through a sequence of steps between both ends. Each step changes the data in some way.

- 1. server reads data from file (or whereever)
- 2. server encrypts data (you add this part)
- 3. server send()s encrypted data

Now the other way around:

- 1. client recv()s encrypted data
- 2. client decrypts data (you add this part)
- 3. client writes data to file (or whereever)

You can also do compression at the same point that you do the encryption/decryption, above. Or you could do both! Just remember to compress before you encrypt. :)

Just as long as the client properly undoes what the server does, the data will be fine in the end no matter how many intermediate steps you add.

So all you need to do to use my code is to find the place between where the data is read and the data is sent (using send()) over the network, and stick some code in there that does the encryption.

What is this "PF_INET" I keep seeing? Is it related to AF_INET?

Yes, yes it is. See the section on socket() for details.

How can I write a server that accepts shell commands from a client and executes them?

For simplicity, lets say the client connect()s, send()s, and close()s the connection (that is, there are no subsequent system calls without the client connecting again.)

The process the client follows is this:

- 1. connect() to server 2. send(''/sbin/ls > /tmp/client.out'')
- 3. close() the connection

Meanwhile, the server is handling the data and executing it:

- 1. accept() the connection from the client
- 2. recv(str) the command string
- 3. close() the connection
- 4. system(str) to run the command

Beware! Having the server execute what the client says is like giving remote shell access and people can do things to your account when they connect to the server. For instance, in the above example, what if the client sends "rm -rf "? It deletes everything in your account, that's what!

So you get wise, and you prevent the client from using any except for a couple utilities that you know are safe, like the **foobar** utility:

```
if (!strcmp(str, "foobar")) {
    sprintf(sysstr, "%s > /tmp/server.out", str);
    system(sysstr);
```

But you're still unsafe, unfortunately: what if the client enters "foobar; rm -rf ~"? The safest thing to do is to write a little routine that puts an escape ("\") character in front of all non-alphanumeric characters (including spaces, if appropriate) in the arguments for the command.

As you can see, security is a pretty big issue when the server starts executing things the client sends.

I'm sending a slew of data, but when I recv(), it only receives 536 bytes or 1460 bytes at a time. But if I run it on my local machine, it receives all the data at the same time. What's going on?

You're hitting the MTU-the maximum size the physical medium can handle. On the local machine, you're using the loopback device which can handle 8K or more no problem. But on ethernet, which can only handle 1500 bytes with a header, you hit that limit. Over a modem, with 576 MTU (again, with header), you hit the even lower limit.

You have to make sure all the data is being sent, first of all. (See the sendall() function implementation for details.) Once you're sure of that, then you need to call recv() in a loop until all your data is read.

Read the section Son of Data Encapsulation for details on receiving complete packets of data using multiple calls to recv().

I'm on a Windows box and I don't have the fork() system call or any kind of struct sigaction. What to do?

If they're anywhere, they'll be in POSIX libraries that may have shipped with your compiler. Since I don't have a Windows box, I really can't tell you the answer, but I seem to remember that Microsoft has a POSIX compatibility layer and that's where fork() would be. (And maybe even sigaction.)

Search the help that came with VC++ for "fork" or "POSIX" and see if it gives you any clues.

If that doesn't work at all, ditch the fork()/sigaction stuff and replace it with the Win32 equivalent: CreateProcess(). I don't know how to use CreateProcess()-it takes a bazillion arguments, but it should be covered in the docs that came with VC++.

How do I send data securely with TCP/IP using encryption?

Check out the OpenSSL project¹⁹.

I'm behind a firewall-how do I let people outside the firewall know my IP address so they can connect to my machine?

Unfortunately, the purpose of a firewall is to prevent people outside the firewall from connecting to machines inside the firewall, so allowing them to do so is basically considered a breach of security.

This isn't to say that all is lost. For one thing, you can still often connect() through the firewall if it's doing some kind of masquerading or NAT or something like that. Just design your programs so that you're always the one initiating the connection, and you'll be fine.

If that's not satisfactory, you can ask your sysadmins to poke a hole in the firewall so that people can connect to you. The firewall can forward to you either through it's NAT software, or through a proxy or something like that.

Be aware that a hole in the firewall is nothing to be taken lightly. You have to make sure you don't give bad people access to the internal network; if you're a beginner, it's a lot harder to make software secure than you might imagine.

Don't make your sysadmin mad at me. ;-)

¹⁹ http://www.openssl.org/

8. Man Pages

In the Unix world, there are a lot of manuals. They have little sections that describe individual functions that you have at your disposal.

Of course, **manual** would be too much of a thing to type. I mean, no one in the Unix world, including myself, likes to type that much. Indeed I could go on and on at great length about how much I prefer to be terse but instead I shall be brief and not bore you with long-winded diatribes about how utterly amazingly brief I prefer to be in virtually all circumstances in their entirety.

[Applause]

Thank you. What I am getting at is that these pages are called "man pages" in the Unix world, and I have included my own personal truncated variant here for your reading enjoyment. The thing is, many of these functions are way more general purpose than I'm letting on, but I'm only going to present the parts that are relevant for Internet Sockets Programming.

But wait! That's not all that's wrong with my man pages:

- They are incomplete and only show the basics from the guide.
- There are many more man pages than this in the real world.
- They are different than the ones on your system.
- The header files might be different for certain functions on your system.
- The function parameters might be different for certain functions on your system.

If you want the real information, check your local Unix man pages by typing man whatever, where "whatever" is something that you're incredibly interested in, such as "accept". (I'm sure Microsoft Visual Studio has something similar in their help section. But "man" is better because it is one byte more concise than "help". Unix wins again!)

So, if these are so flawed, why even include them at all in the Guide? Well, there are a few reasons, but the best are that (a) these versions are geared specifically toward network programming and are easier to digest than the real ones, and (b) these versions contain examples!

Oh! And speaking of the examples, I don't tend to put in all the error checking because it really increases the length of the code. But you should absolutely do error checking pretty much any time you make any of the system calls unless you're totally 100% sure it's not going to fail, and you should probably do it even then!

8.1. accept()

Accept an incoming connection on a listening socket

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
int accept(int s, struct sockaddr *addr, socklen_t *addrlen);
```

Description

Once you've gone through the trouble of getting a SOCK_STREAM socket and setting it up for incoming connections with listen(), then you call accept() to actually get yourself a new socket descriptor to use for subsequent communication with the newly connected client.

The old socket that you are using for listening is still there, and will be used for further accept() calls as they come in.

The listen()ing socket descriptor.

This is filled in with the address of the site that's connecting to you. addr

addrlen

This is filled in with the sizeof() the structure returned in the addr parameter. You can safely ignore it if you assume you're getting a struct sockaddr_in back, which you know you are, because that's the type you passed in for addr.

accept() will normally block, and you can use select() to peek on the listening socket descriptor ahead of time to see if it's "ready to read". If so, then there's a new connection waiting to be accept()ed! Yay! Alternatively, you could set the O_NONBLOCK flag on the listening socket using fcntl(), and then it will never block, choosing instead to return -1 with errno set to EWOULDBLOCK.

The socket descriptor returned by accept() is a bona fide socket descriptor, open and connected to the remote host. You have to close() it when you're done with it.

Return Value

accept() returns the newly connected socket descriptor, or -1 on error, with errno set appropriately.

Example

```
int s, s2;
struct sockaddr_in myaddr, remoteaddr;
socklen_t remoteaddr_len;
myaddr.sin_family = AF_INET;
myaddr.sin_port = htons(3490); // clients connect to this port
myaddr.sin_addr.s_addr = INADDR_ANY; // autoselect IP address
s = socket(PF_INET, SOCK_STREAM, 0);
bind(s, (struct sockaddr*)myaddr, sizeof(myaddr));
listen(s, 10); // set s up to be a server (listening) socket
for(;;) {
    s2 = accept(s, &remoteaddr, &remoteaddr_len);
```

```
// now you can send() and recv() with the // connected client via socket {\tt s2}
}
```

```
socket(), listen(), struct sockaddr_in
```

8.2. bind()

Associate a socket with an IP address and port number

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
int bind(int sockfd, struct sockaddr *my_addr, socklen_t addrlen);
```

Description

When a remote machine wants to connect to your server program, it needs two pieces of information: the IP address and the port number. The bind() call allows you to do just that.

First, you call socket() to get a socket descriptor, and then you load up a struct sockaddr_in with the IP address and port number information, and then you pass both of those into bind(), and the IP address and port are magically (using actual magic) bound to the socket!

If you don't know your IP address, or you know you only have one IP address on the machine, or you don't care which of the machine's IP addresses is used, you can simply set the s_addr field in your struct sockaddr_in to INADDR_ANY and it will fill in the IP address for you.

Lastly, the addrlen parameter should be set to sizeof (my_addr).

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
struct sockaddr_in myaddr;
int s;
myaddr.sin_family = AF_INET;
myaddr.sin_port = htons(3490);
// you can specify an IP address:
inet_aton("63.161.169.137", &myaddr.sin_addr.s_addr);
// or you can let it automatically select one:
myaddr.sin_addr.s_addr = INADDR_ANY;
s = socket(PF_INET, SOCK_STREAM, 0);
bind(s, (struct sockaddr*)myaddr, sizeof(myaddr));
```

```
socket(), struct sockaddr_in, struct in_addr
```

8.3. connect()

Connect a socket to a server

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
int connect(int sockfd, const struct sockaddr *serv_addr,
            socklen_t addrlen);
```

Description

Once you've built a socket descriptor with the socket() call, you can connect() that socket to a remote server using the well-named connect() system call. All you need to do is pass it the socket descriptor and the address of the server you're interested in getting to know better. (Oh, and the length of the address, which is commonly passed to functions like this.)

If you haven't yet called bind() on the socket descriptor, it is automatically bound to your IP address and a random local port. This is usually just fine with you, since you really don't care what your local port is; you only care what the remote port is so you can put it in the serv_addr parameter. You can call bind() if you really want your client socket to be on a specific IP address and port, but this is pretty rare.

Once the socket is connect()ed, you're free to send() and recv() data on it to your heart's content.

Special note: if you connect() a SOCK_DGRAM UDP socket to a remote host, you can use send() and recv() as well as sendto() and recvfrom(). If you want.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
struct sockaddr_in serv_addr;
// pretend the server is at 63.161.169.137 listening on port 80:
myaddr.sin_family = AF_INET;
myaddr.sin_port = htons(80);
inet_aton("63.161.169.137", &myaddr.sin_addr.s_addr);
s = socket(PF_INET, SOCK_STREAM, 0);
connect(s, (struct sockaddr*)myaddr, sizeof(myaddr));
// now we're ready to send() and recv()
```

```
socket(), bind()
```

8.4. close()

Close a socket descriptor

Prototypes

```
#include <unistd.h>
int close(int s);
```

Description

After you've finished using the socket for whatever demented scheme you have concocted and you don't want to send() or recv() or, indeed, do anything else at all with the socket, you can close() it, and it'll be freed up, never to be used again.

The remote side can tell if this happens one of two ways. One: if the remote side calls recv(), it will return 0. Two: if the remote side calls send(), it'll recieve a signal SIGPIPE and send() will return -1 and errno will be set to EPIPE.

Windows users: the function you need to use is called closesocket(), not close(). If you try to use close() on a socket descriptor, it's possible Windows will get angry... And you wouldn't like it when it's angry.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
s = socket(PF_INET, SOCK_DGRAM, 0);
.
// a whole lotta stuff...*BRRRONNNN!*
close(s); // not much to it, really.
```

```
socket(), shutdown()
```

8.5. gethostname()

Returns the name of the system

Prototypes

```
#include <sys/unistd.h>
int gethostname(char *name, size_t len);
```

Description

Your system has a name. They all do. This is a slightly more Unixy thing than the rest of the networky stuff we've been talking about, but it still has its uses.

For instance, you can get your host name, and then call gethostbyname() to find out your IP address.

The parameter name should point to a buffer that will hold the host name, and len is the size of that buffer in bytes. gethostname() won't overwrite the end of the buffer (it might return an error, or it might just stop writing), and it will NUL-terminate the string if there's room for it in the buffer.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
char hostname[128];
gethostname(hostname, sizeof(hostname));
printf("My hostname: %s\n", hostname);
```

See Also

gethostbyname()

8.6. gethostbyname(), gethostbyaddr()

Get an IP address for a hostname, or vice-versa

Prototypes

```
#include <sys/socket.h>
#include <netdb.h>
struct hostent *gethostbyname(const char *name);
struct hostent *gethostbyaddr(const char *addr, int len, int type);
```

Description

These functions map back and forth between host names and IP addresses. After all, you want an IP address to pass to connect(), right? But no one wants to remember an IP address. So you let your users type in things like "www.yahoo.com" instead of "66.94.230.35".

gethostbyname() takes a string like "www.yahoo.com", and returns a struct hostent which contains tons of information, including the IP address. (Other information is the official host name, a list of aliases, the address type, the length of the addresses, and the list of addresses it's a general-purpose structure that's pretty easy to use for our specific purposes once you see how.)

gethostbyaddr() takes a struct in_addr and brings you up a corresponding host name (if there is one), so it's sort of the reverse of gethostbyname(). As for parameters, even though addr is a char*, you actually want to pass in a pointer to a struct in_addr. len should be sizeof(struct in_addr), and type should be AF_INET.

So what is this struct hostent that gets returned? It has a number of fields that contain information about the host in question.

char *h_name	The real canonical host name.
char **h_aliases	A list of aliases that can be accessed with arrays-the last element is NULL
int h_addrtype	The result's address type, which really should be AF_INET for our purposes
int length	The length of the addresses in bytes, which is 4 for IP (version 4) addresses.
char **h_addr_list	A list of IP addresses for this host. Although this is a char**, it's really an array of struct in_addr*s in disguise. The last array element is NULL.
h_addr	A commonly defined alias for h_addr_list[0]. If you just want any old IP address for this host (yeah, they can have more than one) just use this field.

Return Value

Returns a pointer to a resultant struct hostent or success, or NULL on error.

Instead of the normal perror() and all that stuff you'd normally use for error reporting, these functions have parallel results in the variable h_errno, which can be printed using the functions herror() or hstrerror(). These work just like the classic errno, perror(), and strerror() functions you're used to.

Example

```
int i;
struct hostent *he;
struct in_addr **addr_list;
struct in_addr addr;
// get the addresses of www.yahoo.com:
he = gethostbyname("www.yahoo.com");
if (he == NULL) { // do some error checking
    herror("gethostbyname"); // herror(), NOT perror()
    exit(1);
}
// print information about this host:
printf("Official name is: %s\n", he->h_name);
printf("IP address: %s\n", inet_ntoa(*(struct in_addr*)he->h_addr));
printf("All addresses: ");
addr_list = (struct in_addr **)he->h_addr_list;
for(i = 0; addr_list[i] != NULL; i++) {
    printf("%s ", inet_ntoa(*addr_list[i]));
printf("\n");
// get the host name of 66.94.230.32:
inet_aton("66.94.230.32", &addr);
he = gethostbyaddr(&addr, sizeof(addr), AF_INET);
printf("Host name: %s\n", he->h_name);
```

See Also

gethostname(), errno, perror(), strerror(), struct in_addr

8.7. getpeername()

Return address info about the remote side of the connection

Prototypes

```
#include <sys/socket.h>
int getpeername(int s, struct sockaddr *addr, socklen_t *len);
```

Description

Once you have either accept()ed a remote connection, or connect()ed to a server, you now have what is known as a peer. Your peer is simply the computer you're connected to, identified by an IP address and a port. So...

getpeername() simply returns a struct sockaddr_in filled with information about the machine you're connected to.

Why is it called a "name"? Well, there are a lot of different kinds of sockets, not just Internet Sockets like we're using in this guide, and so "name" was a nice generic term that covered all cases. In our case, though, the peer's "name" is it's IP address and port.

Although the function returns the size of the resultant address in len, you must preload len with the size of addr.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
struct sockaddr_in server, addr;
socklen_t len;
// make a socket
s = socket(PF_INET, SOCK_STREAM, 0);
// connect to a server
server.sin_family = AF_INET;
inet_aton("63.161.169.137", &server.sin_addr);
server.sin_port = htons(80);
connect(s, (struct sockaddr*)&server, sizeof(server));
// get the peer name
// we know we just connected to 63.161.169.137:80, so this should print:
//
      Peer IP address: 63.161.169.137
      Peer port
len = sizeof(addr);
getpeername(s, (struct sockaddr*)&addr, &len);
printf("Peer IP address: %s\n", inet_ntoa(addr.sin_addr));
printf("Peer port
                     : %d\n", ntohs(addr.sin_port));
```

```
gethostname(), gethostbyname(), gethostbyaddr()
```

8.8. errno

Holds the error code for the last system call

Prototypes

```
#include <errno.h>
int errno;
```

Description

This is the variable that holds error information for a lot of system calls. If you'll recall, things like socket() and listen() return -1 on error, and they set the exact value of errno to let you know specifically which error occurred.

The header file errno.h lists a bunch of constant symbolic names for errors, such as EADDRINUSE, EPIPE, ECONNREFUSED, etc. Your local man pages will tell you what codes can be returned as an error, and you can use these at run time to handle different errors in different ways.

Or, more commonly, you can call perror() or strerror() to get a human-readable version of the error.

Return Value

The value of the variable is the latest error to have transpired, which might be the code for "success" if the last action succeeded.

Example

```
s = socket(PF_INET, SOCK_STREAM, 0);
if (s == -1)
   perror("socket"); // or use strerror()
tryagain:
if (select(n, &readfds, NULL, NULL) == -1) {
    // an error has occurred!!
    // if we were only interrupted, just restart the select() call:
   if (errno == EINTR) goto tryagain; // AAAA! goto!!!
    // otherwise it's a more serious error:
   perror("select");
   exit(1);
}
```

```
perror(), strerror()
```

8.9. fcntl()

Control socket descriptors

Prototypes

```
#include <sys/unistd.h>
#include <sys/fcntl.h>
int fcntl(int s, int cmd, long arg);
```

Description

This function is typically used to do file locking and other file-oriented stuff, but it also has a couple socket-related functions that you might see or use from time to time.

Parameter s is the socket descriptor you wish to operate on, cmd should be set to F_SETFL, and arg can be one of the following commands. (Like I said, there's more to fcntl() than I'm letting on here, but I'm trying to stay socket-oriented.)

Set the socket to be non-blocking. See the section on blocking for O_NONBLOCK more details.

O_ASYNC

Set the socket to do asynchronous I/O. When data is ready to be recv()'d on the socket, the signal SIGIO will be raised. This is rare to see, and beyond the scope of the guide. And I think it's only available on certain systems.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Different uses of the fcntl() actually have different return values, but I haven't covered them here because they're not socket-related. See your local fcntl() man page for more information.

Example

```
int s = socket(PF_INET, SOCK_STREAM, 0);
fcntl(s, F_SETFL, O_NONBLOCK); // set to non-blocking
fcntl(s, F_SETFL, O_ASYNC);
                                // set to asynchronous I/O
```

See Also

Blocking, send()

```
8.10. htons(), htonl(), ntohs(), ntohl()
```

Convert multi-byte integer types from host byte order to network byte order

Prototypes

```
#include <netinet/in.h>
uint32_t htonl(uint32_t hostlong);
uint16_t htons(uint16_t hostshort);
uint32_t ntohl(uint32_t netlong);
uint16_t ntohs(uint16_t netshort);
```

Description

Just to make you really unhappy, different computers use different byte orderings internally for their multibyte integers (i.e. any interger that's larger than a char.) The upshot of this is that if you send() a two-byte short int from an Intel box to a Mac (before they became Intel boxes, too, I mean), what one computer thinks is the number 1, the other will think is the number 256, and vice-versa.

The way to get around this problem is for everyone to put aside their differences and agree that Motorola and IBM had it right, and Intel did it the weird way, and so we all convert our byte orderings to "big-endian" before sending them out. Since Intel is a "little-endian" machine, it's far more politically correct to call our preferred byte ordering "Network Byte Order". So these functions convert from your native byte order to network byte order and back again.

(This means on Intel these functions swap all the bytes around, and on PowerPC they do nothing because the bytes are already in Network Byte Order. But you should always use them in your code anyway, since someone might want to build it on an Intel machine and still have things work properly.)

Note that the types involved are 32-bit (4 byte, probably int) and 16-bit (2 byte, very likely short) numbers. 64-bit machines might have a htonll() for 64-bit ints, but I've not seen it. You'll just have to write your own.

Anyway, the way these functions work is that you first decide if you're converting from host (your machine's) byte order or from network byte order. If "host", the the first letter of the function you're going to call is "h". Otherwise it's "n" for "network". The middle of the function name is always "to" because you're converting from one "to" another, and the penultimate letter shows what you're converting to. The last letter is the size of the data, "s" for short, or "l" for long. Thus:

```
htons()
              host to network short
htonl()
              host to network long
ntohs()
              network to host short
              network to host long
ntohl()
```

Return Value

Each function returns the converted value.

Example

```
uint32_t some_long = 10;
uint16_t some_short = 20;
```

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```
uint32_t network_byte_order;
// convert and send
network_byte_order = htonl(some_long);
send(s, &network_byte_order, sizeof(uint32_t), 0);
some_short == ntohs(htons(some_short)); // this expression is true
```

8.11. inet_ntoa(), inet_aton()

Convert IP addresses from a dots-and-number string to a struct in_addr and back

Prototypes

```
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
char *inet_ntoa(struct in_addr in);
int inet_aton(const char *cp, struct in_addr *inp);
in_addr_t inet_addr(const char *cp);
```

Description

All of these functions convert from a struct in_addr (part of your struct sockaddr_in, most likely) to a string in dots-and-numbers format (e.g. "192.168.5.10") and viceversa. If you have an IP address passed on the command line or something, this is the easiest way to get a struct in_addr to connect() to, or whatever. If you need more power, try some of the DNS functions like gethostbyname() or attempt a coup-de-tat in your local country.

The function inet_ntoa() converts a network address in a struct in_addr to a dotsand-numbers format string. The "n" in "ntoa" stands for network, and the "a" stands for ASCII for historical reasons (so it's "Network To ASCII"-the "toa" suffix has an analogous friend in the C library called atoi() which converts an ASCII string to an integer.)

The function inet_aton() is the opposite, converting from a dots-and-numbers string into a in_addr_t (which is the type of the field s_addr in your struct in_addr.)

Finally, the function inet_addr() is an older function that does basically the same thing as inet_aton(). It's theoretically deprecated, but you'll see it alot and the police won't come get you if you use it.

Return Value

inet_aton() returns non-zero if the address is a valid one, and it returns zero if the address is invalid.

inet_ntoa() returns the dots-and-numbers string in a static buffer that is overwritten with each call to the function.

inet_addr() returns the address as an in_addr_t, or -1 if there's an error. (That is the same result as if you tried to convert the string "255.255.255", which is a valid IP address. This is why inet_aton() is better.)

Example

```
struct sockaddr_in antelope;
char *some_addr;
inet_aton("10.0.0.1", &antelope.sin_addr); // store IP in antelope
some_addr = inet_ntoa(antelope.sin_addr); // return the IP
printf("%s\n", some_addr); // prints "10.0.0.1"
// and this call is the same as the inet_aton() call, above:
antelope.sin_addr.s_addr = inet_addr("10.0.0.1");
```

```
gethostbyname(), gethostbyaddr()
```

8.12. listen()

Tell a socket to listen for incoming connections

Prototypes

```
#include <sys/socket.h>
int listen(int s, int backlog);
```

Description

You can take your socket descriptor (made with the socket() system call) and tell it to listen for incoming connections. This is what differentiates the servers from the clients, guys.

The backlog parameter can mean a couple different things depending on the system you on, but loosely it is how many pending connections you can have before the kernel starts rejecting new ones. So as the new connections come in, you should be quick to accept() them so that the backlog doesn't fill. Try setting it to 10 or so, and if your clients start getting "Connection refused" under heavy load, set it higher.

Before calling listen(), your server should call bind() to attach itself to a specific port number. That port number (on the server's IP address) will be the one that clients connect to.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
int s;
struct sockaddr_in myaddr;
myaddr.sin_family = AF_INET;
myaddr.sin_port = htons(3490); // clients connect to this port
myaddr.sin_addr.s_addr = INADDR_ANY; // autoselect IP address
s = socket(PF_INET, SOCK_STREAM, 0);
bind(s, (struct sockaddr*)myaddr, sizeof(myaddr));
listen(s, 10); // set s up to be a server (listening) socket
// then have an accept() loop down here somewhere
```

```
accept(), bind(), socket()
```

8.13. perror(), strerror()

Print an error as a human-readable string

Prototypes

```
#include <stdio.h>
void perror(const char *s);
#include <string.h>
char *strerror(int errnum);
```

Description

Since so many functions return -1 on error and set the value of the variable errno to be some number, it would sure be nice if you could easily print that in a form that made sense to you.

Mercifully, perror () does that. If you want more description to be printed before the error, you can point the parameter s to it (or you can leave s as NULL and nothing additional will be printed.)

In a nutshell, this function takes errno values, like ECONNRESET, and prints them nicely, like "Connection reset by peer."

The function strerror() is very similar to perror(), except it returns a pointer to the error message string for a given value (you usually pass in the variable errno.)

Return Value

strerror() returns a pointer to the error message string.

Example

```
int s;
s = socket(PF_INET, SOCK_STREAM, 0);
if (s == -1) { // some error has occurred
    // prints "socket error: " + the error message:
    perror("socket error");
// similarly:
if (listen(s, 10) == -1) {
    // this prints "an error: " + the error message from errno:
    printf("an error: %s\n", strerror(errno));
```

See Also

errno

8.14. poll()

Test for events on multiple sockets simultaneously

Prototypes

```
#include <sys/poll.h>
int poll(struct pollfd *ufds, unsigned int nfds, int timeout);
```

Description

This function is very similar to select() in that they both watch sets of file descriptors for events, such as incoming data ready to recv(), socket ready to send() data to, out-of-band data ready to recv(), errors, etc.

The basic idea is that you pass an array of nfds struct pollfds in ufds, along with a timeout in milliseconds (1000 milliseconds in a second.) The timeout can be negative if you want to wait forever. If no event happen on any of the socket descriptors by the timeout, poll()

Each element in the array of struct pollfds represents one socket descriptor, and contains the following fields:

```
struct pollfd {
   int fd;
                   // the socket descriptor
                   // bitmap of events we're interested in
   short events;
   short revents; // when poll() returns, bitmap of events that occurred
};
```

Before calling poll(), load fd with the socket descriptor (if you set fd to a negative number, this struct pollfd is ignored and its revents field is set to zero) and then construct the events field by bitwise-ORing the following macros:

POLLIN Alert me when data is ready to recv() on this socket.

POLLOUT Alert me when I can send() data to this socket without blocking.

POLLPRI Alert me when out-of-band data is ready to recv() on this socket.

Once the poll() call returns, the revents field will be constructed as a bitwise-OR of the above fields, telling you which descriptors actually have had that event occur. Additionally, these other fields might be present:

POLLERR An error has occurred on this socket.

The remote side of the connection hung up. POLLHUP

POLLNVAL Something was wrong with the socket descriptor fd-maybe it's

uninitialized?

Return Value

Returns the number of elements in the ufds array that have had event occur on them; this can be zero if the timeout occurred. Also returns -1 on error (and errno will be set accordingly.)

Example

```
int s1, s2;
int rv;
char buf1[256], buf2[256];
struct pollfd ufds[2];
```

```
s1 = socket(PF_INET, SOCK_STREAM, 0);
s2 = socket(PF_INET, SOCK_STREAM, 0);
// pretend we've connected both to a server at this point
//connect(s1, ...)...
//connect(s2, ...)...
// set up the array of file descriptors.
^{\prime\prime}/ in this example, we want to know when there's normal or out-of-band
// data ready to be recv()'d...
ufds[0].fd = s1;
ufds[0].events = POLLIN | POLLPRI; // check for normal or out-of-band
ufds[1].events = POLLIN; // check for just normal data
// wait for events on the sockets, 3.5 second timeout
rv = poll(ufds, 2, 3500);
if (rv == -1) {
    perror("poll"); // error occurred in poll()
} else if (rv == 0) {
    printf("Timeout occurred! No data after 3.5 seconds.\n");
} else {
    // check for events on s1:
if (ufds[0].revents & POLLIN) {
         recv(s1, buf1, sizeof(buf1), 0); // receive normal data
    if (ufds[0].revents & POLLPRI) {
    recv(s1, buf1, sizeof(buf1), MSG_OOB); // out-of-band data
    }
     // check for events on s2:
    if (ufds[1].revents & POLLIN) {
   recv(s1, buf2, sizeof(buf2), 0);
    }
}
```

See Also

select()

8.15. recv(), recvfrom()

Recieve data on a socket

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
ssize_t recv(int s, void *buf, size_t len, int flags);
ssize_t recvfrom(int s, void *buf, size_t len, int flags,
                 struct sockaddr *from, socklen_t *fromlen);
```

Description

Once you have a socket up and connected, you can read incoming data from the remote side using the recv() (for TCP SOCK_STREAM sockets) and recvfrom() (for UDP SOCK_DGRAM

Both functions take the socket descriptor s, a pointer to the buffer buf, the size (in bytes) of the buffer len, and a set of flags that control how the functions work.

Additionally, the recvfrom() takes a struct sockaddr*, from that will tell you where the data came from, and will fill in fromlen with the size of struct sockaddr. (You must also initialize fromlen to be the size of from or struct sockaddr.)

So what wonderous flags can you pass into this function? Here are some of them, but you should check your local man pages for more information and what is actually supported on your system. You bitwise-or these together, or just set flags to 0 if you want it to be a regular vanilla recv().

MSG_OOB

Recieve Out of Band data. This is how to get data that has been sent to you with the MSG_OOB flag in send(). As the recieving side, you will have had signal SIGURG raised telling you there is urgent data. In your handler for that signal, you could call recv() with this MSG_00B flag.

MSG_PEEK

If you want to call recv() "just for pretend", you can call it with this flag. This will tell you what's waiting in the buffer for when you call recv() "for real" (i.e. without the MSG_PEEK flag. It's like a sneak preview into the next recv() call.

MSG_WAITALL

Tell recv() to not return until all the data you specified in the len parameter. It will ignore your wishes in extreme circumstances, however, like if a signal interrupts the call or if some error occurs or if the remote side closes the connection, etc. Don't be mad with it.

When you call recv(), it will block until there is some data to read. If you want to not block, set the socket to non-blocking or check with select() or poll() to see if there is incoming data before calling recv() or recvfrom().

Return Value

Returns the number of bytes actually recieved (which might be less than you requested in the len paramter), or -1 on error (and errno will be set accordingly.)

If the remote side has closed the connection, recv() will return 0. This is the normal method for determining if the remote side has closed the connection. Normality is good, rebel!

Example

```
int s1, s2;
int byte_count, fromlen;
struct sockaddr_in addr;
char buf [512];
// show example with a TCP stream socket first
s1 = socket(PF_INET, SOCK_STREAM, 0);
// info about the server
addr.sin_family = AF_INET;
addr.sin_port = htons(3490);
inet_aton("10.9.8.7", &addr.sin_addr);
connect(s1, &addr, sizeof(addr)); // connect to server
// all right! now that we're connected, we can recieve some data!
byte_count = recv(s1, buf, sizeof(buf), 0);
printf("recv()'d %d bytes of data in buf\n", byte_count);
// now demo for UDP datagram sockets:
s2 = socket(PF_INET, SOCK_DGRAM, 0);
fromlen = sizeof(addr);
byte_count = recvfrom(s2, buf, sizeof(buf), 0, &addr, &fromlen);
printf("recv()'d %d bytes of data in buf\n", byte_count);
printf("from IP address %s\n", inet_ntoa(addr.sin_addr));
```

```
send(), sendto(), select(), poll(), Blocking
```

8.16. select()

Check if sockets descriptors are ready to read/write

Prototypes

```
#include <sys/select.h>
int select(int n, fd_set *readfds, fd_set *writefds,
           fd_set *exceptfds, struct timeval *timeout);
FD_SET(int fd, fd_set *set);
FD_CLR(int fd, fd_set *set);
FD_ISSET(int fd, fd_set *set);
FD_ZERO(fd_set *set);
```

Description

The select() function gives you a way to simultaneously check multiple sockets to see if they have data waiting to be recv()d, or if you can send() data to them without blocking, or if some exception has occurred.

You populate your sets of socket descriptors using the macros, like FD_SET(), above. Once you have the set, you pass it into the function as one of the following parameters: readfds if you want to know when any of the sockets in the set is ready to recv() data, writefds if any of the sockets is ready to send() data to, and/or exceptfds if you need to know when an exception (error) occurs on any of the sockets. Any or all of these parameters can be NULL if you're not interested in those types of events. After select() returns, the values in the sets will be changed to show which are ready for reading or writing, and which have exceptions.

The first parameter, n is the highest-numbered socket descriptor (they're just ints, remember?) plus one.

Lastly, the struct timeval, timeout, at the end-this lets you tell select() how long to check these sets for. It'll return after the timeout, or when an event occurs, whichever is first. The struct timeval has two fields: tv_sec is the number of seconds, to which is added tv_usec, the number of microseconds (1,000,000 microseconds in a second.)

The helper macros do the following:

```
FD_SET(int fd, fd_set *set);
                                         Add fd to the set.
FD_CLR(int fd, fd_set *set);
                                          Remove fd from the set.
                                         Return true if fd is in the set.
FD_ISSET(int fd, fd_set *set);
                                         Clear all entries from the set.
FD_ZERO(fd_set *set);
```

Return Value

Returns the number of descriptors in the set on success, 0 if the timeout was reached, or -1 on error (and errno will be set accordingly.) Also, the sets are modified to show which sockets are ready.

Example

```
int s1, s2, n;
fd_set readfds;
struct timeval tv;
char buf1[256], buf2[256];
```

```
s1 = socket(PF_INET, SOCK_STREAM, 0);
s2 = socket(PF_INET, SOCK_STREAM, 0);
// pretend we've connected both to a server at this point
//connect(s1, ...)...
//connect(s2, ...)...
// clear the set ahead of time
FD_ZERO(&readfds);
// add our descriptors to the set
FD_SET(s1, &readfds);
FD_SET(s2, &readfds);
// since we got s2 second, it's the "greater", so we use that for
// the n param in select()
n = s2 + \bar{1};
// wait until either socket has data ready to be recv()d (timeout 10.5 secs)
tv.tv_sec = 10;
tv.tv_usec = 500000;
rv = select(n, &readfds, NULL, NULL, &tv);
if (rv == -1) {
    perror("select"); // error occurred in select()
} else if (rv == 0) {
    printf("Timeout occurred! No data after 10.5 seconds.\n");
} else {
    // one or both of the descriptors have data
    if (FD_ISSET(s1, &readfds)) {
        recv(s1, buf1, sizeof(buf1), 0);
    if (FD_ISSET(s2, &readfds)) {
        recv(s1, buf2, sizeof(buf2), 0);
    }
}
```

See Also

poll()

8.17. setsockopt(), getsockopt()

Set various options for a socket

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
int getsockopt(int s, int level, int optname, void *optval,
               socklen_t *optlen);
int setsockopt(int s, int level, int optname, const void *optval,
               socklen_t optlen);
```

Description

Sockets are fairly configurable beasts. In fact, they are so configurable, I'm not even going to cover it all here. It's probably system-dependent anyway. But I will talk about the basics.

Obviously, these functions get and set certain options on a socket. On a Linux box, all the socket information is in the man page for socket in section 7. (Type: "man 7 socket" to get all these goodies.)

As for parameters, s is the socket you're talking about, level should be set to SOL_SOCKET. Then you set the optname to the name you're interested in. Again, see your man page for all the options, but here are some of the most fun ones:

SO_BINDTODEVICE	Bind this socket to a symbolic device name like eth0 instead
	of using bind() to bind it to an IP address. Type the command
	ifconfig under Unix to see the device names.

SO_REUSEADDR Allows other sockets to bind() to this port, unless there is

an active listening socket bound to the port already. This enables you to get around those "Address already in use" error messages

when you try to restart your server after a crash.

Allows UDP datagram (SOCK_DGRAM) sockets to send and SO_BROADCAST recieve packets sent to and from the broadcast address. Does

nothing-NOTHING!!-to TCP stream sockets! Hahaha!

As for the parameter optval, it's usually a pointer to an int indicating the value in question. For booleans, zero is false, and non-zero is true. And that's an absolute fact, unless it's different on your system. If there is no parameter to be passed, optval can be NULL.

The final parameter, optlen, is filled out for you by getsockopt() and you have to specify it for getsockopt(), where it will probably be sizeof(int).

Warning: on some systems (notably Sun and Windows), the option can be a char instead of an int, and is set to, for example, a character value of '1' instead of an int value of 1. Again, check your own man pages for more info with "man setsockopt" and "man 7 socket"!

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
int optval;
int optlen;
char *optval2;
```

```
// set SO_REUSEADDR on a socket to true (1): optval = \overline{1};
setsockopt(s1, SOL_SOCKET, SO_REUSEADDR, &optval, sizeof(optval));
// bind a socket to a device name (might not work on all systems):
optval2 = "eth1"; // 4 bytes long, so 4, below:
setsockopt(s2, SOL_SOCKET, SO_BINDTODEVICE, optval2, 4);
// see if the SO_BROADCAST flag is set:
getsockopt(s3, SOL_SOCKET, SO_BROADCAST, &optval, &optlen);
if (optval != 0) {
    print("SO_BROADCAST enabled on s3!\n");
```

See Also

fcntl()

8.18. send(), sendto()

Send data out over a socket

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
ssize_t send(int s, const void *buf, size_t len,
             int flags);
ssize_t sendto(int s, const void *buf, size_t len,
               int flags, const struct sockaddr *to,
               socklen_t tolen);
```

Description

These functions send data to a socket. Generally speaking, send() is used for TCP SOCK_STREAM connected sockets, and sendto() is used for UDP SOCK_DGRAM unconnected datagram sockets. With the unconnected sockets, you must specify the destination of a packet each time you send one, and that's why the last parameters of sendto() define where the packet is going.

With both send() and sendto(), the parameter s is the socket, buf is a pointer to the data you want to send, len is the number of bytes you want to send, and flags allows you to specify more information about how the data is to be sent. Set flags to zero if you want it to be "normal" data. Here are some of the commonly used flags, but check your local send() man pages for more details:

MSG_OOB	Send as "out of band" data. TCP supports this, and it's a way
	to tell the receiving system that this data has a higher priority than
	the normal data. The receiver will recieve the signal SIGURG and
	it can then recieve this data without first recieving all the rest of
	the normal data in the queue.

MSG_DONTROUTE Don't send this data over a router, just keep it local.

MSG_DONTWAIT If send() would block because outbound traffic is clogged,

> have it return EAGAIN. This is like a "enable non-blocking just for this send." See the section on blocking for more details.

MSG_NOSIGNAL If you send() to a remote host which is no longer recv()ing,

you'll typically get the signal SIGPIPE. Adding this flag prevents

that signal from being raised.

Return Value

Returns the number of bytes actually sent, or -1 on error (and errno will be set accordingly.) Note that the number of bytes actually sent might be less than the number you asked it to send! See the section on handling partial send()s for a helper function to get around this.

Also, if the socket has been closed by either side, the process calling send() will get the signal SIGPIPE. (Unless send() was called with the MSG_NOSIGNAL flag.)

Example

```
int spatula_count = 3490;
char *secret_message = "The Cheese is in The Toaster";
```

```
int stream_socket, dgram_socket;
struct sockaddr_in dest;
int temp;
// first with TCP stream sockets:
stream_socket = socket(PF_INET, SOCK_STREAM, 0);
!
// convert to network byte order
temp = htonl(spatula_count);
// send data normally:
send(stream_socket, &temp, sizeof(temp), 0);
// send secret message out of band:
send(stream_socket, secret_message, strlen(secret_message)+1, MSG_OOB);
// now with UDP datagram sockets:
dgram_socket = socket(PF_INET, SOCK_DGRAM, 0);
// build destination
dest.sin_family = AF_INET;
inet_aton("10.0.0.1", &dest.sin_addr);
dest.sin_port = htons(2223);
// send secret message normally:
sendto(dgram_socket, secret_message, strlen(secret_message)+1, 0,
       (struct sockaddr*)&dest, sizeof(dest));
```

See Also

recv(), recvfrom()

8.19. shutdown()

Stop further sends and recieves on a socket

Prototypes

```
#include <sys/socket.h>
int shutdown(int s, int how);
```

Description

That's it! I've had it! No more send()s are allowed on this socket, but I still want to recv() data on it! Or vice-versa! How can I do this?

When you close() a socket descriptor, it closes both sides of the socket for reading and writing, and frees the socket descriptor. If you just want to close one side or the other, you can use this shutdown() call.

As for parameters, s is obviously the socket you want to perform this action on, and what action that is can be specified with the how paramter. How can be SHUT_RD to prevent further recv()s, SHUT_WR to prohibit further send()s, or SHUT_RDWR to do both.

Note that shutdown() doesn't free up the socket descriptor, so you still have to eventually close() the socket even if it has been fully shut down.

This is a rarely used system call.

Return Value

Returns zero on success, or -1 on error (and errno will be set accordingly.)

Example

```
int s = socket(PF_INET, SOCK_STREAM, 0);
// ...do some send()s and stuff in here...
// and now that we're done, don't allow any more sends()s:
shutdown(s, SHUT_RD);
```

See Also

close()

8.20. socket()

Allocate a socket descriptor

Prototypes

```
#include <sys/types.h>
#include <sys/socket.h>
int socket(int domain, int type, int protocol);
```

Description

Returns a new socket descriptor that you can use to do sockety things with. This is generally the first call in the whopping process of writing a socket program, and you can use the result for subsequent calls to listen(), bind(), accept(), or a variety of other functions.

domain

domain describes what kind of socket you're interested in. This can, believe me, be a wide variety of things, but since this is a socket guide, it's going to be PF_INET for you. And, correspondingly, when you load up your struct sockaddr_in to use with this socket, you're going to set the sin_family field to AF_INET

(Also of interest is PF_INET6 if you're going to be doing IPv6 stuff. If you don't know what that is, don't worry about it...yet.)

type

Also, the type parameter can be a number of things, but you'll probably be setting it to either SOCK_STREAM for reliable TCP sockets (send(), recv()) or SOCK_DGRAM for unreliable fast UDP sockets (sendto(), recvfrom().)

(Another interesting socket type is SOCK_RAW which can be used to construct packets by hand. It's pretty cool.)

protocol

Finally, the protocol parameter tells which protocol to use with a certain socket type. Like I've already said, for instance, SOCK_STREAM uses TCP. Fortunately for you, when using SOCK_STREAM or SOCK_DGRAM, you can just set the protocol to 0, and it'll use the proper protocol automatically. Otherwise, you can use getprotobyname() to look up the proper protocol number.

Return Value

The new socket descriptor to be used in subsequent calls, or -1 on error (and errno will be set accordingly.)

Example

```
int s1, s2;
s1 = socket(PF_INET, SOCK_STREAM, 0);
s2 = socket(PF_INET, SOCK_DGRAM, 0);
if (s1 == -1) {
     perror("socket");
```

```
accept(), bind(), listen()
```

8.21. struct sockaddr_in, struct in_addr

Structures for handling internet addresses

Prototypes

```
#include <netinet/in.h>
struct sockaddr_in {
                    sin_family;
                                 // e.g. AF_INET
   short
   unsigned short sin_port; // e.g. htons(3490)
   struct in_addr sin_addr;
                                 // see struct in_addr, below
                    sin_zero[8]; // zero this if you want to
   char
};
struct in_addr {
   unsigned long s_addr; // load with inet_aton()
```

Description

These are the basic structures for all syscalls and functions that deal with internet addresses. In memory, the struct sockaddr_in is the same size as struct sockaddr, and you can freely cast the pointer of one type to the other without any harm, except the possible end of the universe.

Just kidding on that end-of-the-universe thing...if the universe does end when you cast a struct sockaddr_in* to a struct sockaddr*, I promise you it's pure coincidence and you shouldn't even worry about it.

So, with that in mind, remember that whenever a function says it takes a struct sockaddr* you can cast your struct sockaddr_in* to that type with ease and safety.

There's also this sin_zero field which some people claim must be set to zero. Other people don't claim anything about it (the Linux documentation doesn't even mention it at all), and setting it to zero doesn't seem to be actually necessary. So, if you feel like it, set it to zero using memset().

Now, that struct in_addr is a weird beast on different systems. Sometimes it's a crazy union with all kinds of #defines and other nonsense. But what you should do is only use the s_addr field in this structure, because many systems only implement that one.

With IPv4 (what basically everyone in 2005 still uses), the struct s_addr is a 4-byte number that represents one digit in an IP address per byte. (You won't ever see an IP address with a number in it greater than 255.)

Example

```
struct sockaddr_in myaddr;
int s;
myaddr.sin_family = AF_INET;
myaddr.sin_port = htons(3490);
inet_aton("63.161.169.137", &myaddr.sin_addr.s_addr);
s = socket(PF_INET, SOCK_STREAM, 0);
bind(s, (struct sockaddr*)myaddr, sizeof(myaddr));
```

```
accept(), bind(), connect(), inet_aton(), inet_ntoa()
```

9. More References

You've come this far, and now you're screaming for more! Where else can you go to learn more about all this stuff?

9.1. Books

For old-school actual hold-it-in-your-hand pulp paper books, try some of the following excellent guides. Note the prominent Amazon.com logo. What all this shameless commercialism means is that I basically get a kickback (Amazon.com store credit, actually) for selling these books through this guide. So if you're going to order one of these books anyway, why not send me a special thank you by starting your spree from one of the links, below.

Besides, more books for me might ultimately lead to more guides for you. ;-)



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Unix Network Programming, volumes 1-2 by W. Richard Stevens. Published by Prentice Hall. ISBNs for volumes 1-2: 013490012X²¹, 0130810819²².

Internetworking with TCP/IP, volumes I-III by Douglas E. Comer and David L. Stevens. Published by Prentice Hall. ISBNs for volumes I, II, and III: 0130183806²³, 0139738436²⁴, 0138487146^{25} .

TCP/IP Illustrated, volumes 1-3 by W. Richard Stevens and Gary R. Wright. Published by Addison Wesley. ISBNs for volumes 1, 2, and 3: 020163346926, 020163354X27, 020163495328.

TCP/IP Network Administration by Craig Hunt. Published by O'Reilly & Associates, Inc. ISBN 1565923227²⁹.

Advanced Programming in the UNIX Environment by W. Richard Stevens. Published by Addison Wesley. ISBN 020156317730.

Using C on the UNIX System by David A. Curry. Published by O'Reilly & Associates, Inc. ISBN 0937175234. Out of print.

9.2. Web References

On the web:

BSD Sockets: A Quick And Dirty Primer³¹ (has other great Unix system programming info, too!)

The Unix Socket FAQ³²

```
^{20}~{\tt http://www.amazon.com/exec/obidos/redirect-home/beejsguides-20}
21 http://www.amazon.com/exec/obidos/ASIN/013490012X/beejsguides-20
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31 http://www.cs.umn.edu/~bentlema/unix/
32 http://www.developerweb.net/sock-faq/
```

Client-Server Computing³³
Intro to TCP/IP³⁴
Another Different Intro to TCP/IP³⁵
TCP/IP FAQ³⁶
The Winsock FAQ³⁷

9.3. RFCs

RFCs³⁸–the real dirt:

RFC-768³⁹-The User Datagram Protocol (UDP)

RFC-791⁴⁰—The Internet Protocol (IP)

RFC-793⁴¹–The Transmission Control Protocol (TCP)

*RFC-854*⁴²—The Telnet Protocol

*RFC-951*⁴³–The Bootstrap Protocol (BOOTP)

*RFC-1350*⁴⁴–The Trivial File Transfer Protocol (TFTP)

³³ http://pandonia.canberra.edu.au/ClientServer/

³⁴ http://pclt.cis.yale.edu/pclt/COMM/TCPIP.HTM

 $^{^{35} \ \}mathtt{http://www.doc.ic.ac.uk/~ih/doc/pc_conn/tcpip/intro/intro0.html}$

 $^{^{36} \ \}mathtt{http://www.faqs.org/faqs/internet/tcp-ip/tcp-ip-faq/part1/}$

 $^{^{37} \; \}texttt{http://tangentsoft.net/wskfaq/}$

³⁸ http://www.rfc-editor.org/

³⁹ http://www.rfc-editor.org/rfc/rfc768.txt

⁴⁰ http://www.rfc-editor.org/rfc/rfc791.txt

⁴¹ http://www.rfc-editor.org/rfc/rfc793.txt

⁴² http://www.rfc-editor.org/rfc/rfc854.txt

⁴³ http://www.rfc-editor.org/rfc/rfc951.txt

⁴⁴ http://www.rfc-editor.org/rfc/rfc1350.txt