

(This part is a review from the lecture)

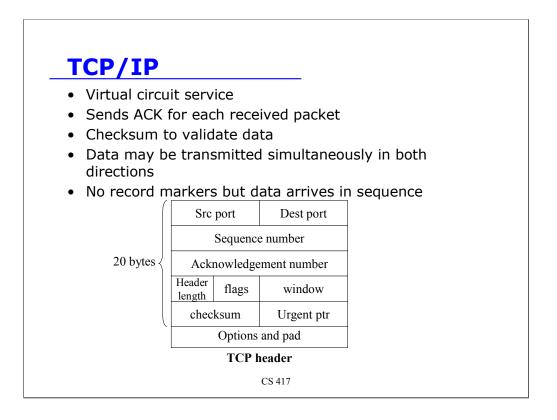
The IP protocol allows packets to be sent from one machine to another machine. Each IP packet header identifies the source and destination machines by their 32-bit IP address.

The assumption at the IP layer is that the delivery of packets is unreliable (packets may get lost or data may get corrupted) and may not be in-order (packets may arrive in a different order than they were sent.

Two <u>transport-level</u> protocols allow us to deal with application-to-application communication. These are TCP/IP and UDP/IP.

Both TCP and UDP use **port numbers** to identify transport endpoints (applications). A port number is a 16-bit number that is associated with an application.

For example, if Process A (which grabbed use of port number 1512) communicates with Process B (which grabbed the use of port 25), an IP packet from process A to process B will be addressed with a source address of 192.168.1.5 and a destination address of 192.168.1.7. The TCP or UDP header, which is encapsulated *within* the IP data will contain a source port of 1512 and a destination port of 25.



TCP/IP – Transport Control Protocol

Provides virtual circuit service – connection oriented – you set up a connection and just read and write data on that connection. No need to address each packet.

Packets are acknowledged by the protocol - no chance of packet loss

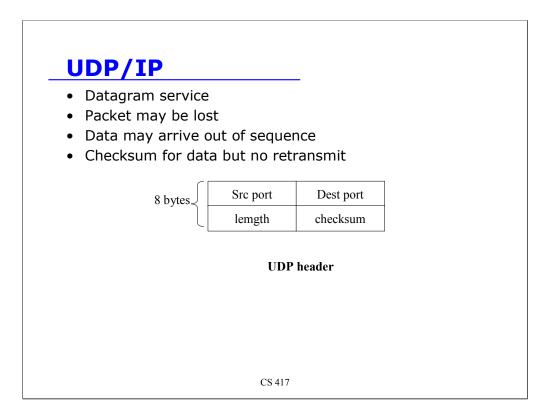
A checksum is present to validate data – no data corruption. If there is, an acknowledgement is not sent and the packet is retransmitted

TCP does not maintain record boundaries. For example, if you perform three writes: 5 bytes, 20 bytes, 15 bytes, the other side may receive all the data in a single read of 40 bytes. Or... it may not. If you want to preserve these boundaries, it is up to you to layer your own protocol on top of this. For example:

send (length, data), (length, data), ...

Instead of

data, data, data, ...



UDP/IP - User Datagram Protocol

Provides datagram service – connectionless– each packet must be addressed to the destination

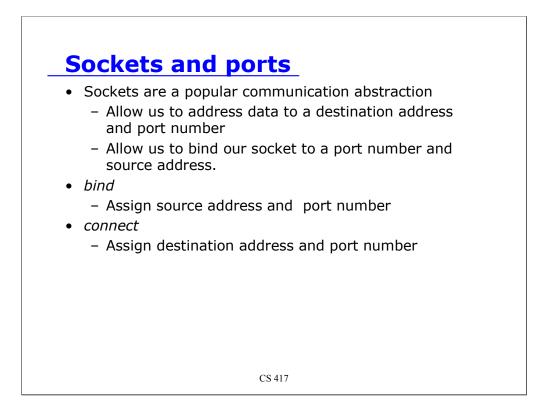
Data may be lost: it is up to the application to discover that a packet did not arrive at its destination. Application can send acknowledgements if this is needed.

Data may arrive out of sequence: if packets take a different route via IP they may arrive in a different order from the one sent. TCP buffers up out-of-order packets in the kernel and presents them in the right order to the user.

A checksum for the data exists in the UDP header. If data corruption is detected then the packet is dropped. To the application, this is the same as a lost packet.

UDP is a lighter weight protocol: header is 8 bytes vs. 20+ bytes for TCP. Linux source for udp is around 1200 lines vs. >8000 lines for tcp.

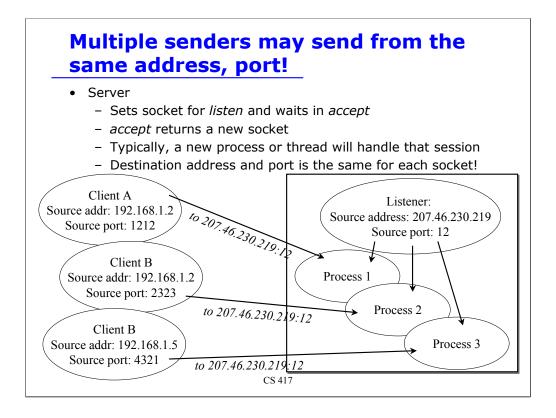
More importantly, dealing with UDP traffic uses less kernel resources: state about the connection, receive buffers to deal with out-of-order data, retransmission does not have to be handled.



As we've seen with sockets programs, sockets are a popular implementation for achieving network communication.

After creating a socket, we use the <u>bind</u> system call to assign it to an address and port number. Typically, this address is 0.0.0.0, which represents INADDR\_ANY, or any valid incoming address (particularly useful if a machine has several connections and several IP addresses). For servers, the port number is usually chosen by the user so that other programs will be able to specify it. For clients, one typically asks the operating system to pick any unused port number (this is requested by setting the port to 0 in *bund*).

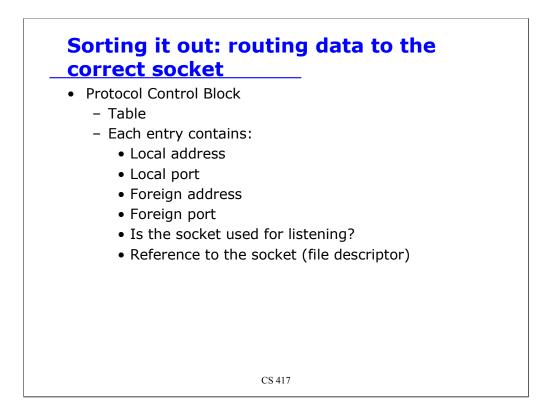
For connection-oriented protocols (TCP), the server address and port number is specified with the *connect* system call. For connectionless protocols (UDP), the address and port number may be specified explicitly when sending data using *sendto* or *sendmsg* system calls.



When a server set up a socket for listening and waits for an incoming connection with the *accept* call, a new socket is returned for each accepted connection.

A server will typically create a new thread or fork a new process that will be responsible for handling that particular communication session.

Each client still addresses data to the same IP address and port number. Now that data has to be routed to the correct socket so that the right process or thread can get the data and all the data streams do not become interleaved.



How does the operating system get incoming data to the correct socket when multiple sockets may have the same incoming address and port number?

The operating system maintains a structure called the **Protocol Control Block** (PCB).

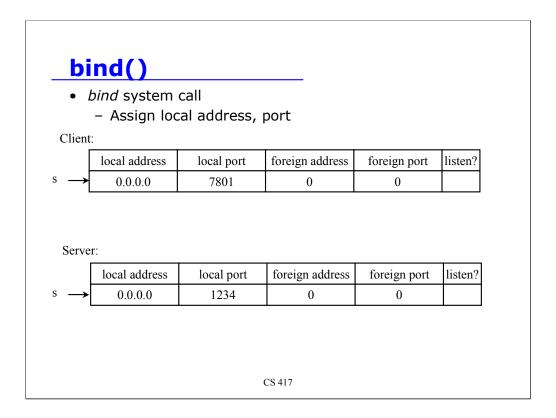
The PCB is a table of entries, where each PCB entry contains the following information:

- local address: IP address bound to the socket
- local port: port number bound to the socket
- foreign address: IP address from the remote side
- foreign port: port number from the remote side
- is the socket used for listening?
- reference to the socket descriptor (file descriptor)

Each socket descriptor contains a pointer to an entry in the PCB table.

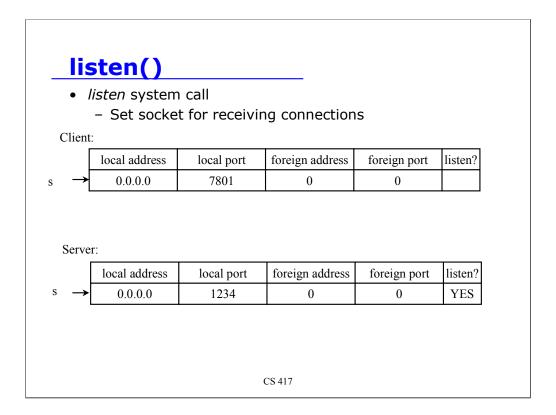
<ul> <li>socket syster</li> <li>Allocate a</li> </ul>		in PCB table		
ient:				
local address	local port	foreign address	foreign port	listen?
$\rightarrow 0$	0	0	0	
new entry in				
erver:				
	local port	foreign address	foreign port	listen?
erver:		foreign address 0	foreign port 0	listen?

The first thing we do in using sockets is to allocate a new socket with the *socket* system call. This causes a new, empty, entry to be created in the PCB table.



The bind system call assigns a source address to a socket. This causes the local address and local port fields in the corresponding PCB entry (for that socket) to be filled in.

Suppose the client binds to address 0.0.0.0 (INADDR\_ANY), port 7801 Suppose the server binds to address 0.0.0.0 (INADDR\_ANY), port 1234



The *listen* system call is used by a server to set a socket for receiving future *connect* requests. This causes a flag to be set in the PCB entry for that socket

	onnect(				
•	<i>connect</i> syste – Send a <i>co</i>		est to server		
Client	: send a <i>connect</i> c	control message	[from 135.250.68.4	3, port 7801]	
	local address	local port	foreign address	foreign port	listen?
		7001	0	0	
→	0.0.0.0	7801	0	0	
Serve			<i>ccept</i> , creates new		
Serve			1		listen?
	er: waits for contro	ol messages in ad	ccept, creates new	PCB entry	listen? YES
Serve $5 $ Snew $$	er: waits for contro local address	ol messages in <i>ac</i> local port	<i>ccept</i> , creates new foreign address	PCB entry foreign port	

The *connect* system call is used by a client to establish a connection to the server.

The client gets the remote address and port as part of the address structure passed to *connect*.

It then sends a *connect* control message to the server. The server should be waiting in *accept* to receive this message.

When the server receives a control message,

it scans the PCB table for an entry where the connect request matches the local address (0 matches anything) and the socket is in the listen state (I.e., set to receive control messages).

- a new socket is allocated and a new PCB entry is created that is associated with the new socket.

- this new PCB entry is populated with the local address on which the connection was received and the foreign address, which is the source address of the *connect* request.

•	<i>connect</i> syste – Send a <i>co</i>		est to server		
Clien	t: send a <i>connect</i> c	ontrol message	[from 135.250.68.4	3, port 7801]	
	local address	local port	foreign address	foreign port	listen?
<b>→</b>	0.0.0.0	7801	192.11.35.15	1234	
→ Serve			192.11.35.15		
→ Serve			1 1		listen?
$\rightarrow$ Serve	er: waits for contro	l messages in ad	ccept, creates new I	PCB entry	listen? YES

The server sends a message to the client acknowledging acceptance of the connection. This message contains the real address and port of the server that the client puts in its PCB entry for that socket.

Each message from the client is tagged either as a **data** message or a **control** message (e.g. connect).

If the message is a data message:

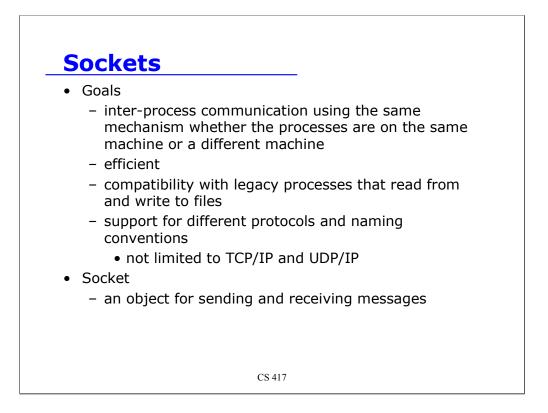
search through the list of PCB entries for one where the foreign address and foreign port match the incoming message

If the message is a control message:

search through the list of PCB entries for one where *listen* is set (and foreign address = foreign port = 0).

# **Programming with sockets**

CS 417 Distributed Systems Design

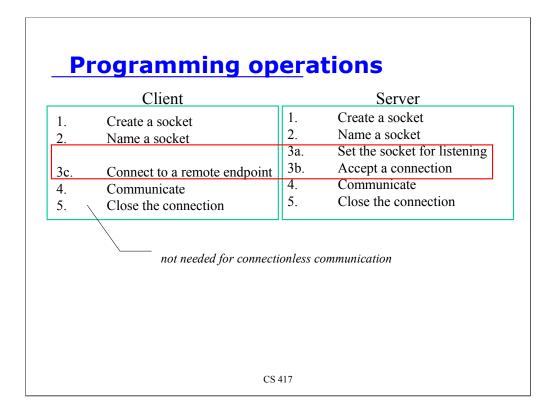


Sockets are an attempt at creating a generalized IPC (inter-process communication) model.

They were developed by Berkeley in 1982 as part of the Berkeley version of Unix (BSD 4.1a, with general availability in BSD 4.2 in September 1983).

The design goals were:

- have a uniform mechanism at the operating system level for inter-process communication, independent of whether the processes are on the same machine.
- the interface should be as efficient as possible
- any applications that do not know about networking or IPC should be able to be given input and output file descriptors and just use them without changes for network communication.
- For example another process may accept or initiate the communication,
- create the "file" descriptors, and then hand off the descriptors to a process
- (e.g., by fork & exec) that is network ignorant
- the sockets mechanism should work for a just about any naming scheme and networking protocol it is not meant only for TCP/IP and UDP/IP.



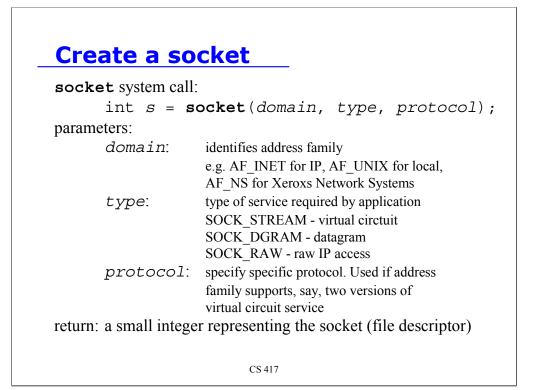
The system-call level interface consists of the following operations:

- 1. Both the client and server need to create a socket.
- 2. Since the socket identifies the *transport endpoint* for the application, the transport address will need to be assigned this is known as *naming* the socket. For IP communication, this means assigning a port number to the socket.
- 3a. A server has to indicate that it wants to accept connections on this socket.
- 3b. A server will then block (go to sleep), waiting for incoming connections.
- 3c. A client will connect to a server.
- 4. Each party will communicate by sending and receiving messages.
- 5. When done, each party will close the connection.

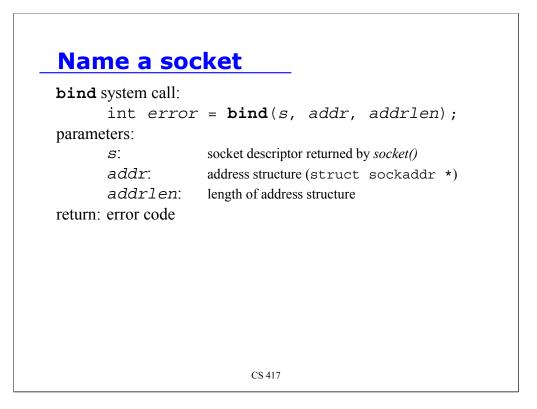
For connectionless protocols, steps 3a-c are not needed. A client can just send a message to a server and a server can block while waiting to receive a message. It is up to the server now to disambiguate messages from different client.

# Unix system call interface

- System calls:
  - socket
  - bind
  - listen
  - accept
  - connect
  - read/write, send/recv, sendto/recvfrom, sendmsg/recvmsg
  - close/shutdown







The bind system call allows us to assign a transport endpoint to the socket.

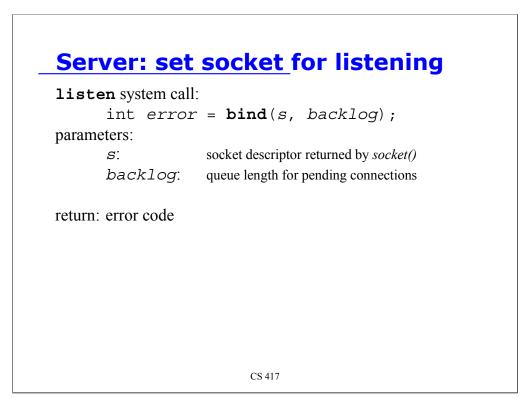
Analogy: socket: request a phone line

bind: request a phone number for the line

For IP services, this is where we specify the port number on which we want to accept connections. Since the client will not be accepting connections, it generally will not care what port number it gets, so it can specify a port of 0, in which case the operating system will assign it an available port number.

Note - the client needs a port number because messages from the server back to the client have to be addressed as well.

**bind** is a separate system call because some communication domains may not require it. Also, some communication domains may require other custom operations to take place on the socket before the binding is performed.



The *listen* system call is used by the server to indicate that it will accept connection requests on that socket.

Once done, the operating system will accept such connection requests on behalf of this socket.

The server will generally be sleeping, waiting for a connection request (via the *accept* system call - next slide). When it gets the connection, it will usually fork off another process or create/dispatch a thread to handle that request and go back to listening for more connections. Since there is a delay between accepting a connection request and looping back to accept another one, the *listen* system call allows one to specify a **backlog** - a queue size for pending connections.

# Server: accept connections

accept system call:

int snew = accept(s, clntaddr, addrlen);
parameters:

 s:
 socket descriptor returned by socket()

 clntaddr:
 struct sockaddr \*

 contains returned client address information

 addrlen:
 length of address information

return: a new socket to be used for this communication session

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The *accept* system call blocks (by default) and waits until a connection comes in on the socket.

Accept creates a **new socket** on which this communication session will take place. The original socket, created with the *socket* system call on the server, is used *only* to get connections.

In addition to returning a new socket, *accept* also returns information about the address of the connecting client (could be used for authentication checking...).

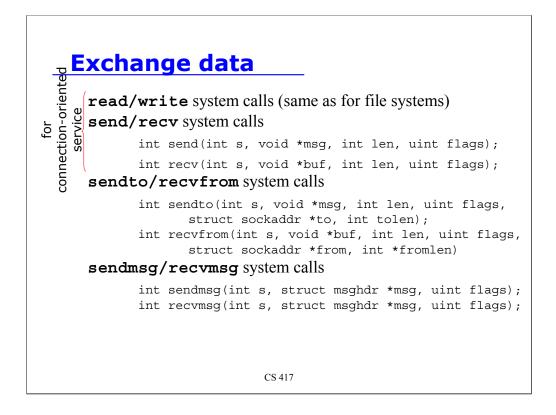
connect	system cal	1:			
in	t error	= con	nect(s)	svraddr,	addrlen)
parameters	3:				
s:		socket d	escriptor retu	med by socket()	
$_{SV}$	raddr:	struct	sockaddı	<u>*</u>	
		contains	address of se	rver	
ad	drlen:	length of	f address info	rmation	
return: err	or code				

The *connect* system call is used by the client to connect to the server.

s - socket returned by the socket() system call

svraddr - structure containing the address of the server. The contents depend on the transport being used

addrlen - length of the address structure. This is passed in because C does not support polymorphism and the address structure length is a function of the transport used.



Data can now be exchanged between client and server.

The client will treat its socket, *s*, as a file descriptor and use ordinary *read/write* system calls on it.

The server will treat its communication socket, *snew*, as a file descriptor and also use ordinary *read/write* system calls on it.

Alternatively, the client and server may choose to use **send/recv** system calls, which support flags that may be useful for certain communication transports, such as:

- process out of band data
- bypass routing use direct interface
- don't block
- don't generate a SIGPIPE signal when either end breaks the connection

For **connectionless service**, *accept* and *listen* were not invoked by the server and *connect* was not invoked by the client. The server just receives messages from anyone who sends them. The system calls **sendto/recvfrom** and **sendmsg/recvmsg** allow addresses to be specified on the send line, along with other information. [see man pages]

# Stop all further communication

file-sytem **close** system call:

```
close(s);
```

or

shutdown system call:

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

where

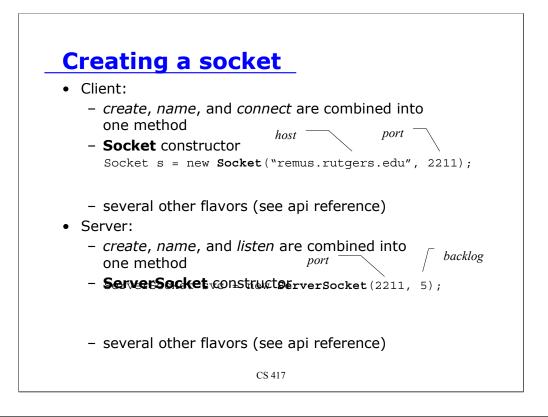
s: socket
how: 0: further receives disallowed
1: further sends disallowed
2: further sends and receives disallowed

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A socket can be closed with the normal file-system-based **close** system call or, for greater control, the **shutdown** call can be used. This allows one to disallow certain types of operations (sends or receives).

# Using sockets in java

- java.net package
  - Socket class
    - deals with sockets used for communication
  - ServerSocket class
    - deals with sockets used for accepting connections
  - DatagramSocket class
    - deals with datagram packets
  - both Socket and ServerSocket rely on the SocketImpl class to actually implement sockets







### • accept method of ServerSocket

- block until connection arrives
- return a **Socket**

```
ServerSocket svc = new ServerSocket(2211, 5);
Socket req = svc.accept();
```

# Exchange data

- Obtain InputStream and OutputStream from Socket
   layer whatever you need on top of them
  - e.g. DataInputStream, PrintStream, BufferedReader, ...

Example:

#### <u>client</u>

```
DataInputStream in = new DataInputStream(s.getInputStream());
PrintStream out = new PrintStream(s.getOutputStream());
```

### <u>server</u>

### **Close the sockets**

• Close input and output streams first:

### <u>client:</u>

```
try {
    out.close();
    in.close();
    s.close();
} catch (IOException e) {}
```

### server:

```
try {
    out.close();
    in.close();
    req.close(); // close connection socket
    svc.close(); // close ServerSocket
} catch (IOException e) {}
```

```
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```

# Sample programs

 $http://www.cs.rutgers.edu/{\sim}pxk/rutgers/src/socketdemo.tar$ 

run:

tar xvf SocketDemo.tar will create two directories: jsocketdemo: java sample program socketdemo: C sample program

each directory contains a README file.