

Internet Programming

Programming with Sockets

Brief Introduction to Networks

Introduction

- There is one way computers can communicate
 - By sending network messages to each other
 - All kinds of communication are built from messages
- There is one way programs can send/receive network messages
 - Through sockets
 - All other communication paradigms are built from sockets

Two Different Kinds of Networks

Circuit switching

- One electrical circuit assigned per communication
- Example: the (analog) phone network
- Guaranteed constant quality of service
- Waste of resources (periods of silence), fault tolerance

Packet switching

- Messages are split into packets, which are transmitted independently
 - Packets can take different routes
 - Network infrastructures are shared among users
- Example: the Internet, and most computer networks
- Good resource usage, fault tolerance
- Variable QOS, packets may be delivered in the wrong order

Internet Protocol

- Most computer networks use the Internet Protocol
 - Even if they are not connected to the Internet
- The base protocol: IP (Internet Protocol)
 - Send packets of limited size
 - □ Up to 65,536 bytes
 - But if the MTU (Maximum Transmission Unit) of some link on the path is lower, the packet will be fragmented (IPv4) or dropped (IPv6)
 - Minimum allowed MTU is 576 bytes; in practice nowadays higher
 - Each packet is sent to an IP address
 - **Example:** 130.37.193.13
 - IP offers no guarantee:
 - Packets may get lost
 - Packets may be delivered twice
 - □ Packets may be delivered in wrong order
 - Packets may be corrupted during transfer
- Usually, programs do not use IP directly
 - All other Internet Protocols are built over IP

UDP: User Datagram Protocol

- □ UDP is very similar to IP
 - Send/receive packets
 - No guarantee
- In UDP, packets are called datagrams
 - Each datagram is sent to an IP address and a port number
 - Example: 130.37.193.13 port=1234
- Ports allow to distinguish between several programs running simultaneously on the same machine
 - Program A uses port 1234
 - Program B uses port 1235
 - When a datagram is received, the OS knows which program it should be delivered to.

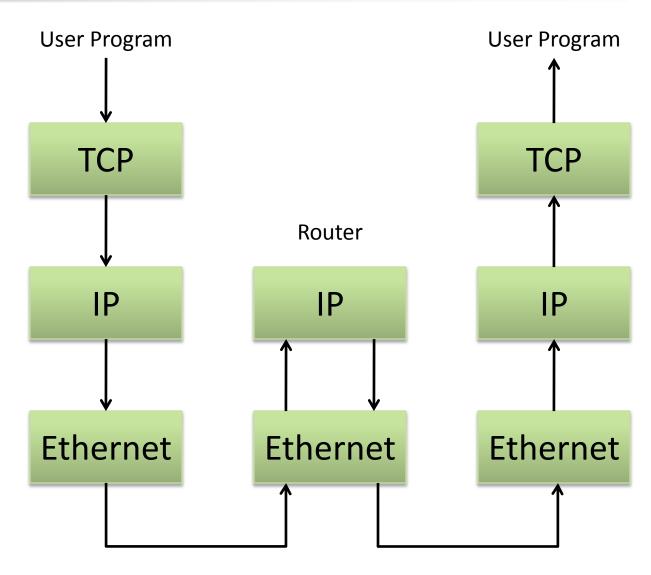
TCP: Transmission Control Protocol

- TCP establishes connections between pairs of machines
 - To communicate with a remote host, we must first connect to it
- □ TCP provides the illusion of a reliable data flow to the users
 - Flows are split into packets, but the users don't see them
 - Provides flow control and congestion control
 - QUESTION: What is the difference?!?
- TCP guarantees that the data sent will not be lost, unordered or corrupted
 - The sender gives numbers to packets so that the receiver can reorder them
 - The receiver acknowledges received packets so that the sender can retransmit lost packets
- Communication is bidirectional
 - The same connection can be used to send data in both directions
 - E.g., a request and its response

A Very Simplified View

- Establish connections
- Split data streams into packets
- ACK received packets
- Retransmit lost packets
- Reorder packets
- Find how to reach a given host
- If on the same network → easy
- Otherwise → send to a router

- Transform packets into electrical signals
- Make sure signals do not interfere with each other



IP Addressing and Name Resolution

IP Address Conversion

- IP Addresses
 - 32-bit integers: **2183468070** (good for computers!)
 - Dotted strings: 130.37.20.38 (good for humans!)
 - DNS name: www.cs.vu.nl (even better for humans!)
- You can convert between integer and dotted string:

- in_addr_t: unsigned 32-bit integer
- struct in_addr: structure containing an in_addr_t:

```
struct in_addr {
   in_addr_t s_addr;
};
```

□ there are historic reasons why this is done that way...

Big/Little-endian, Network Ordering

- Computers represent numbers in different orderings:
 - Is significant byte first or last?
 - □ Big-endian: $0x12345678 \rightarrow 0x12 0x34 0x56 0x78$
 - examples: PowerPC, Sparc, UltraSparc
 - □ Little-endian: $0x12345678 \rightarrow 0x78 0x56 0x34 0x12$
 - examples: Alpha, i386, AMD64
- Network byte ordering
 - A standard representation (defined as Big-Endian)
- To convert numbers: host <---> network ordering

```
#include <netinet/in.h>
uint16_t htons(uint16_t value);    /* Host to Network, 16 bits */
uint32_t htonl(uint32_t value);    /* Host to Network, 32 bits */
uint16_t ntohs(uint16_t value);    /* Network to Host, 16 bits */
uint32_t ntohl(uint32_t value);    /* Network to Host, 32 bits */
```

sockaddr in: Unix Network Addresses

- Unix represents network addresses with a struct sockaddr
 - This structure is generic for all kinds of networks
 - For Internet addresses, we use sockaddr_in

```
struct sockaddr_in {
  sa_family_t    sin_family; /* set to AF_INET */
  in_port_t    sin_port; /* Port number */
  struct in_addr sin_addr; /* Contains the IP address */
};

struct in_addr {
  in_addr_t    s_addr; /* IP address in network ordering */
};
```

- sin_family: indicates which type of address. Always set to AF_INET.
- **sin_port**: port number, in network byte order
- sin_addr.s_addr: IP address, in network byte order. To represent an unspecified IP address, set it to htonl (INADDR ANY).
 - □ QUESTION: When is this useful?

Domain Names

- Internet Protocols are all based on IP addresses
 - But IP addresses are hard for humans to remember
 - Our web server: http://130.37.20.20
 - Better: http://www.cs.vu.nl
- Using Domain Names
 - Domain names cannot be used directly by network protocols
 - Network protocols only use IP addresses
 - But you can convert domain names into IP addresses thanks to DNS
- Domain Name Service (DNS): handles Domain Name resolution
 - Hundreds of thousands of servers around the world that cooperate to resolve addresses
 - To learn more on how this works, go to the Distributed Systems course!

Converting Domain Names to IP

□ Conversion is done by gethostbyname()

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

...where struct hostent is as follows

■ h_addr_list: A null-terminated array of network addresses for the host

gethostbyname()

Example:

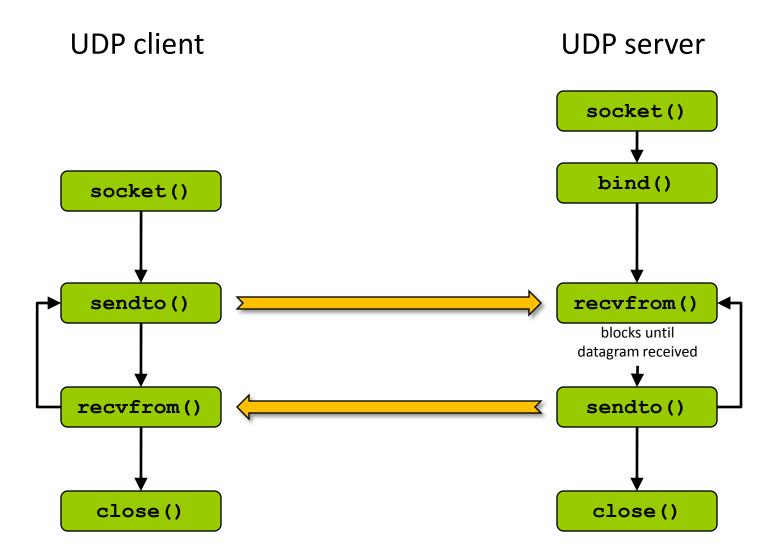
```
#include <netdb.h>
int print resolv(const char *name) {
  struct hostent *resolv;
  struct in addr *addr;
 resolv = gethostbyname(name);
  if (resolv==NULL) {
   printf("Address not found for %s\n", name);
   return -1;
 else {
    addr = (struct in addr*) resolv->h addr list[0];
    printf("The IP address of %s is %s\n", name, inet ntoa(*addr));
    return 0;
```

UDP Sockets

UDP

- Defined in RFC 768
- Popular UDP-based protocols
 - **DNS** Domain Name System
 - **NFS** Network File System
 - **SNMP** Simple Network Management Protocol
 - DHCP Dynamic Host Configuration Protocol
 - RIP Routing Information Protocol

UDP Socket Functions



Creating a Socket

socket() creates a new socket (for either UDP or TCP)

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

- To create an Internet socket, use:
 - domain = AF INET
 - type = SOCK DGRAM for UDP, SOCK STREAM for TCP
 - protocol = 0
 - Return value: socket descriptor, or -1 for error

bind(): Assign an Address to a Socket

- □ bind() is used to specify the address of the socket
 - IP addr + port number

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

- sockfd: the socket descriptor
- my_addr: a pointer to struct sockaddr_in (containing the address)
- addrlen: the size of struct sockaddr_in
- **Return value**: 0 for success, -1 for error
- □ If you don't specify them, the system gives them a value:
 - IP address → the IP address of the running host (this is usually correct)
 - Port number → any number (of an unused port)
- QUESTION: When do you need to specify a port number, and when can you omit it?

Example use of socket() and bind()

■ To create a UDP socket on port 1234:

```
int fd, err;
struct sockaddr_in addr;

fd = socket(AF_INET,SOCK_DGRAM,0);
if (fd<0) { ... }

addr.sin_family = AF_INET;
addr.sin_port = htons(1234);
addr.sin_addr.s_addr = htonl(INADDR_ANY);

err = bind(fd, (struct sockaddr *) &addr, sizeof(struct sockaddr_in));
if (err<0) { ... }</pre>
```

- □ For historic reasons, you are obliged to explicitly cast your struct sockaddr in * into a struct sockaddr *
- **QUESTION:** Is INADDR_ANY equivalent to 127.0.0.1?

Sending and Receiving Datagrams

sendto() and recvfrom() are very similar:

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

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

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

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

sockfd	socket descriptor	
*buf	buffer of message to send/receive	
len	buffer length	
flags	0	0
*to / *from	destination address	source address (value-return)
tolen / *fromlen	sizeof(struct sockaddr)	sizeof() (value-return)
Return value	Number of bytes sent/received, or -1 for error	

Use of sendto() and recvfrom()

Example:

```
(the socket is already created)
char msg[64];
int err:
struct sockaddr in dest;
strcpy(msg, "hello, world!");
dest.sin family = AF INET;
                    = htons(1234);
dest.sin port
dest.sin addr.s addr =
            inet addr("130.37.193.13");
err = sendto(fd,
     msg,
      strlen(msq)+1,
     0,
      (struct sockaddr*) &dest,
      sizeof(struct sockaddr in));
if (err<0) { ... }
```

```
(the socket is created and bound to a
well-known port)
char msq[64];
int len, flen;
struct sockaddr in from;
flen = sizeof(struct sockaddr in);
len = recvfrom(fd,
      msq,
      sizeof(msq),
      0,
      (struct sockaddr*) &from,
      &flen);
if (len<0) { ... }
printf("Received %d bytes from host %s
port %d: %s", len,
       inet ntoa(from.sin addr),
       ntohs(from.sin port),
       msq);
```

Closing a socket

■ Sockets must be closed when they are no longer in use:

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

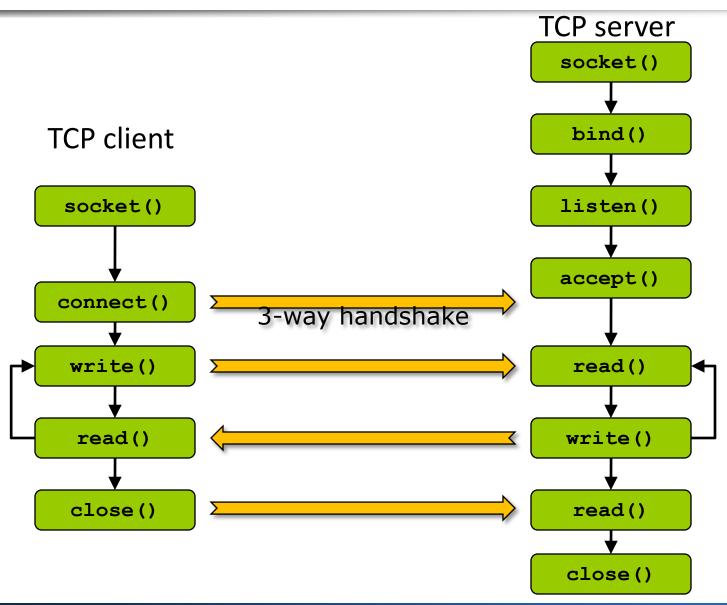
- **sockfd**: the socket descriptor
- **Return value**: 0 for success, -1 for error

TCP Sockets

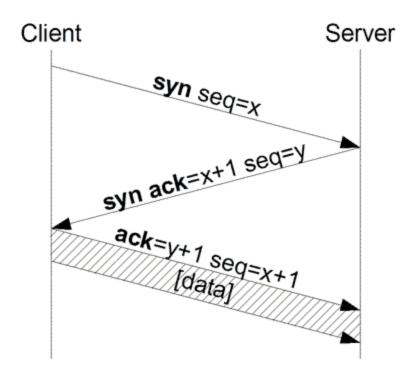
TCP

- Defined in RFC 793
- Popular TCP-based protocols
 - TELNET
 - FTP File Transfer Protocol
 - SMTP Simple Mail Transfer Protocol
 - HTTP Hyper Text Transfer Protocol
 - SSH Secure Shell

TCP Socket Functions



The TCP three-way handshake



Creating a Socket

- Some functions are the same as in UDP
- socket(): creates a socket

```
sockfd = socket(AF INET, SOCK STREAM, 0);
```

- bind(): to specify the address of a socket
 - Only useful for server sockets
 - Exactly like UDP sockets

listen(): Setting a Server Socket

- By default, TCP sockets are created as client sockets
 - A client socket cannot receive incoming connections
- Server sockets need to maintain more state
 - TCP establishes connections thanks to the three-way handshake:
 - ☐ The client sends a connection request
 - The server answers
 - The client acknowledges the server's answer
 - Server sockets must allocate resources for handling connections
- To convert a client socket to a server socket, use listen()
 - And indicate how many not-yet-accepted connections can be supported in parallel
 - If this number is exceeded, the server will refuse connections

listen()

■ The interface is simple

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

- **sockfd**: the socket descriptor
- backlog: the size of the buffer (often set to 5)
- **Return value**: 0 for success, -1 for error

Note

- backlog is <u>not</u> a limit on the number of connections established in parallel!
- It only limits the number of **pending connections** (i.e., connections before having been accepted)

Initiating a TCP connection

Clients initiate connections to servers thanks to connect():

- **sockfd**: the socket descriptor
- serv_addr: a pointer to a struct sockaddr_in containing the address where to connect to
 - Obviously you must specify the destination IP address and port number
- addrlen: sizeof(struct sockaddr_in)
- **Return value**: 0 for success, -1 for error
- connect() binds the client's socket to a random unused port
- **QUESTION:** When can it fail?

Possible Connection Errors

- Destination Unreachable
 - The SYN message receives an ICMP error from some intermediate router
 - The client keeps trying for some time
 - After some time, it returns an EHOSTUNREACH error to the process
- Server reached, but no process bound to that port
 - The SYN message receives a RST (Reset) reply from the server's TCP
 - The client's TCP returns an ECONNREFUSED error to the process instantly
- Server reached, but listen backlog exhausted
 - Like the previous case
- No response → timeout
 - The SYN message receives no response
 - The client's TCP sends SYN message again (after 6sec, 24sec)
 - After some time, an ETIMEDOUT error is returned to the client process

Waiting for an Incoming Connection

- accept() blocks the process until an incoming connection is received
 - When a connection is received, accept() creates a new socket dedicated to this connection
 - The new socket is used to communicate with the client
 - □ The original socket is immediately ready to wait for other connections

accept():

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

- sockfd: the socket descriptor
- addr: a pointer to a sockaddr_in structure where the address of the client will be copied
- **addrlen**: a pointer to an integer containing the size of addr
- Return value: the descriptor of the newly created socket, or -1 for error

Example use of accept()

```
int sock, newsock, res;
sockaddr in client addr;
socklen t addrlen;
(the socket sock is created and bound)
res = listen(sock,5);
if (res < 0) { ... }
addrlen = sizeof(struct sockaddr in);
newsock = accept(sock, (struct sockaddr *) &client addr, &addrlen);
if (newsock < 0) { ... }
else
  printf("Connection from %s!\n", inet ntoa(client addr.sin addr));
}
```

Writing data to a socket

write works the same for sending data to a TCP socket or writing to a file

```
#include <unistd.h>
ssize_t write(int sockfd, const void *buf, size_t count);
```

- sockfd: socket descriptor
- **buf**: buffer to be sent
- count: size of buffer
- Return value: number of bytes sent, or -1 for error
- Attention: When writing to a socket, write may send fewer bytes than requested
 - Due to limits in internal kernel buffer space
- Always check the return value of write, and resend the nontransmitted data

writen()

■ A wrapper function (from Stevens book, page 78)

```
ssize t writen(int fd, const void *vptr, size t n)
{
  size t nleft;
 ssize t nwritten;
 const char *ptr;
 ptr = vptr;
 nleft = n;
 while (nleft > 0) {
    if ( ((nwritten = write(fd, ptr, nleft)) <= 0 ) {</pre>
      if (errno == EINTR)
        nwritten = 0; /* and call write() again */
      else
        return -1; /* error */
    nleft -= nwritten;
    ptr += nwritten;
 return n;
```

Reading data from a socket

read() blocks the process until receiving data from the socket

```
#include <unistd.h>
ssize_t read(int sockfd, void *buf, size_t count);
```

- sockfd: socket descriptor
- **buf**: buffer where to write the data read
- **count**: size of buffer
- Return value: number of bytes read, or -1 for error
- Attention: When reading from a socket, read() may read fewer bytes than requested
 - It delivers the data that have been received
 - This does not mean that the stream of data is finished, there may be more to come
 - The end-of-file (EOF) is notified to the read by read() returning 0

Closing a TCP socket

To stop sending data to a socket, use close():

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

- Anyone can call this, either the client or the server
- This sends an EOF message to the other party
 - When receiving an EOF, read returns 0 bytes
 - Subsequent reads and writes will return errors

Asymmetric Disconnection

■ Sometimes you may want to tell the other party that you are finished, but let it finish before closing the connection

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

- how: SHUT_WR for stopping writing, SHUT_RD for stopping reading
- When one party has closed the connection, the other can still write data (and then close the connection as well)

To initiate a disconnection □ shutdown(fd, SHUT_WR) □ Keep on reading the last data □ Until receiving an EOF □ close() the socket □ To receive a disconnection □ read() keeps receiving data □ read() receives EOF □ Keep on writing the last data □ close() the socket

Socket Options

Socket Options

- Various attributes that are used to determine the behavior of sockets.
- Setting options tells the OS/Protocol Stack the behavior we want.
- Support for generic options (apply to all sockets) and protocol specific options.

Option types

- Many socket options are **boolean flags** indicating whether some feature is enabled (1) or disabled (0).
- Other options are associated with more complex types including int, timeval, in_addr, sockaddr, etc.
- Some options are read-only (we can't set the value)

Getting and Setting option values

Use getsockopt() and setsockopt():

- sockfd: the socket descriptor
- level: SOL_SOCKET (for socket options)
- optname: the option name (a #define)
- **optval**: a buffer containing the option value
- optlen: the length of optval

General Options

- Protocol independent options
- Handled by the generic socket system code
- Some general options are supported only by specific types of sockets (SOCK_DGRAM, SOCK_STREAM)

Some Generic Options

- SO_BROADCAST
- SO_DONTROUTE
- □ SO_ERROR
- SO_KEEPALIVE
- □ SO_LINGER
- SO_RCVBUF,SO_SNDBUF
- SO_REUSEADDR

SO_BROADCAST

- Boolean option: enables/disables sending of broadcast messages
- Underlying Data Link layer must support broadcasting!
- Applies only to SOCK_DGRAM sockets
- Prevents applications from sending broadcasts by mistake
 - OS looks for this flag when broadcast address is specified
 - e.g., ping -b 130.37.193.255

SO_DONTROUTE

- Boolean option: enables bypassing of normal routing
 - If destination is directly attached to an interface, OK.
 - Otherwise, ENETUNREACH is returned
- Used by routing daemons (routed, gated, etc.)
 - Bypasses the routing table
 - The routing table could be incorrect

SO_ERROR

- Integer value option
 - The value is an error indicator value (similar to errno)
 - Read only!
- Reading (by calling getsockopt()) clears any pending error
 - so, you can only read it once

SO_KEEPALIVE

- Boolean option
 - enabled means that TCP sockets should send a probe to peer if no data flow for a "long time"
 - Allows a process to determine whether peer process/host has crashed
- QUESTION: Consider what would happen to an open telnet or ssh connection without keepalive
- Typically used by servers
 - some sort of garbage collection of terminated clients

SO_LINGER

■ Value is of type:

```
#include <sys/socket.h>
struct linger {
  int l_onoff;    /* 0 = off */
  int l_linger;    /* time in seconds */
};
```

- Controls whether and how long a call to close() waits for pending ACKs
 - I_onoff = 0 → default behavior: close() returns immediately, and the system tries to deliver any pending packets from the send buffer
 - $I_onoff = 1 \rightarrow close()$ returns when:
 - either all pending packets have been sent and acknowledged by the remote
 TCP stack
 - or l_linger seconds have elapsed
- Only for connection-oriented sockets (e.g., TCP)

SO_LINGER

- Without SO_LINGER, the node closing a connection has no way of knowing that the other peer received all sent data
- With SO_LINGER, it can know that all sent data reached the other peer's TCP stack
- QUESTION: Isn't this what shutdown(fd, SHUT_WR) does?
 - Not exactly!
 - shutdown(fd, SHUT_WR) closes your own write channel, but you can still read
 - Reading till you receive EOF (read() returns 0 bytes) means that the other peer did a close()
 - Therefore, shutdown() assures you that the other peer has read all your data, while SO_LINGER assures you that your data reached the other peer's TCP stack, but you don't know if it was read!

SO_RCVBUF, SO_SNDBUF

- Integer values options
 - Change the receive and send buffer sizes.
- Can be used with STREAM and DGRAM sockets.
- With TCP, this option affects the window size used for flow control
 - Must be established before the connection is made

SO_REUSEADDR

- Boolean option
 - Enables binding to an address (port) that is already in use.
- Used by servers that are transient
 - Allows binding a passive socket to a port currently in use (with active sockets) by other processes.
- Can be used to establish separate servers for the same service on different interfaces (or different IP addresses on the same interface)
 - Virtual Web Servers can work this way
- Very useful in your assignments!

I/O Multiplexing

I/O Multiplexing

- We saw so far how a process can handle a single connection
 - **accept()** and **read()** block until something is received on a given socket
- How can a program handle multiple sockets?
 - Use multiple processes
 - Use non-blocking I/O
 - □ It works for read() but not for accept()
- select() monitors multiple file descriptors
 - It blocks the process until any one of them is ready for reading or writing
- poll() is similar to select()
 - With additional information about streams

select() [1/3]

select() monitors several file descriptors simultaneously

- n: the highest numbered file descriptor, plus 1
- readfds: a list of file descriptors to monitor for reading
- writefds: a list of file descriptors to monitor for writing
- exceptfds: a list of file descriptors to monitor for exceptions
- timeout: a duration after which select() returns anyway.
 - Set timeout.tv_sec = timeout.tv_usec = 0 for zero timeout
 (return immediately)
 - □ Set timeout = NULL for no timeout
- Return value: the number of (i.e., how many) descriptors ready for I/O, or 0 in case of timeout, or -1 for error

select() [2/3]

- fd_set is a bitset representing a list of file descriptors
 - Do not manipulate fd_set directly, always use special macros:

- select() modifies the contents of readfds, writefds, and exceptfds
- After select()
 - file descriptors that are ready for use are turned on (in their set)
 - non-ready descriptors are turned off
- To wait for a socket to be ready to accept(), put it in the read set

select() [3/3]

Example use:

```
int nb, fd1=5, fd2=8;
char buf[1024];
fd set read set;
while (1) {
  FD ZERO(&read set);
  FD SET(fd1, &read set);
  FD SET(fd2, &read set);
  nb = select(20, &read set, NULL, NULL, NULL);
  if (nb<=0) { ... }
  if (FD ISSET(fd1, &read set)) {
   bzero(buf, 1024);
    nb = read(fd1, buf, 1024);
    if (nb<0) { ... }
    if (nb==0) printf("Received EOF on fd1!\n");
    else printf("Received data on fd1: %s\n", buf);
  if (FD ISSET(fd2, &read set)) { ... }
```

Server Structures

Server Structures

- Often, a server accepts connections to one (TCP) socket
 - But it wants to process several requests in parallel
 - Better use of server's resources
 - Incoming requests can start being executed immediately after reception (not having to wait for previous client)
- Depending on its nature, a server can receive between 0 and dozens of thousands requests per second
- Several server structures can be used:
 - Iterative (i.e., not concurrent)
 - One Process Per Client
 - Preforking
 - select() loop
 - Many other variants...

Iterative Servers

■ An iterative server treats one request after the other

```
int fd, newfd;
while (1) {
  newfd = accept(fd, ...);
  treat_request(newfd);
  close(newfd);
}
```

- Simple
- Potentially low resource utilization
 - If treat_request() does not utilize all the CPU, resources are being wasted
- Potentially long queue of incoming connections waiting for accept()
 - Increased service latency
 - If the queue increases, the server may start rejecting new connections

One Process Per Client [1/2]

- A new child process is created to handle each connection
 - Also known as "One Child Per Client"

```
void sig chld(int) {
 while (waitpid(0, NULL, WNOHANG) > 0) {}
  signal(SIGCHLD, sig chld);
}
int main() {
  int fd, newfd, pid;
 // socket(), bind(), and listen() have been omitted
  signal(SIGCHLD, sig chld);
 while (1) {
   newfd = accept(fd, ...);
    if (newfd < 0) continue;
   pid = fork();
    if (pid==0) { treat request(newfd); exit(0); }
    else { close(newfd); }
```

One Process Per Client [2/2]

- ☐ This is the most common type of concurrent server
 - Several requests are treated simultaneously
 - Incoming requests are accepted and treated immediately
- QUESTION: What are the downsides?
 - It may not be suitable for highly loaded servers
 - □ Ok for ~1K connections per day, but for ~1M?
 - fork() takes a lot of time!
- MORE QUESTIONS!
 - Why is the signal() call necessary?
 - What happens if treat_request() needs to modify a global variable? How can you obtain the desired effect?
 - e.g., to update request statistics

Preforking

- The server first creates a pool of processes dedicated to treating requests
- Each client is delegated to a child process
- □ No fork() is done per connecting client
- Performance gain!

Preforking: Each child calls accept()

Typical example

```
#define NB PROC 10
void recv requests(int fd) { /* An iterative server */
  int newfd;
 while (1) {
   newfd = accept(fd,...);
   treat request(newfd);
    close(newfd);
int main() {
  int fd=socket(AF INET, SOCK STREAM, 0);
  // bind() and listen() have been omitted
  for (int i=0; i<NB PROC; i++) /* Create NB PROC children */
    if (fork()==0)
      recv requests(fd);
```

Preforking: Each child calls accept()

- Multiple accepts on a single socket descriptor
 - In System V types of Unix (Solaris, HP-UX, etc.), not possible!
 - However, allowed in BSD-based Unix systems
- Where it is allowed, accept() internally defines a condition variable
 - Condition: "Block until a client opens a connection"
 - Predicate: "Is a client available in the queue so I can accept it?"
- QUESTION: Can you guess some problem?
 - When a client appears, all processes waiting on that condition variable are woken up
 - Only one is allowed to run at a time, so the first one accepts the client
 - When other processes get the mutex, they first check if the predicate is still valid, and block again since the client is already accepted
 - For large process pools → performance penalty per connecting client

Preforking: Mutex on accept()

A solution is to protect access to accept() by a single mutex shared by all processes:

```
void recv_requests(int fd) { /* An iterative server */
int f;
while (1)
{
    /* --- ACQUIRE MUTEX --- */
    newfd=accept(fd,...);
    /* --- RELEASE MUTEX --- */

    treat_request(newfd);
    close(f);
}
```

- For systems NOT supporting concurrent accepts on a single socket:
 - Solves the problem
- For the rest systems:
 - Improves performance

Preforking: accept() on parent

- Another alternative:
 - accept on the parent
 - pass socket descriptor to a free child
- Difficulties?
 - Keeping track of which children are free or loaded
 - Communicating to the children
 - Keeping open pipes
 - □ Using shared memory, semaphores, condition variables
 - Requires **descriptor passing** (out of the scope of this course)
- Advantages?
 - Parent has control over which processes to delegate clients to
 - Can achieve more efficient memory usage by using the same processes

One Thread Per Client

- Very similar to One Process Per Client
 - Attention to pay with race conditions, due to shared memory!
- Starting a thread per client is faster than preforking a pool of processes!

Prethreading

- Very similar to preforking
- Even faster than One Thread Per Client (which is already faster than Preforking)
- Locking access to accept() by a mutex, further improves performance

Forking vs. Threading

- □ If threading is faster than forking, why not always use threading?
 - The system might not support threads
 - The application might be easier to design as separate processes
 - In <u>certain</u> cases, One Process Per Client can be <u>faster</u> than One Thread Per Client
 - QUESTION: When?! ©
 - **ANSWER:** When the code treating a client needs to execute another program. Then, forking a unithreaded process and then doing an exec would be faster than creating a new thread, then forking a multithreaded process and finally doing the exec

Select Loop

- A single process (and single thread) manages multiple connections thanks to select()
 - This is quite difficult to do correctly
 - You must split request treatment into a set of non-blocking stages
 - You must maintain a list of data structures containing the current state of each concurrent request
 - Which stage it is in
 - All internal data it needs
- Implementing a select loop server is left to the students as exercise
- For example, the Squid Web cache is implemented as a select loop

inetd

- For servers with **low load**, it is recommended to use
 - One Process Per Client
 - ...or even Iterative
- □ For servers with extremely low load (e.g., one invocation per day or week)
 - inetd
 - No need to run the server at all, while not needed
 - Define the service's protocol (TCP/UDP), port, and executable in /etc/services
 - inetd listens on that port
 - Upon reception of a client, it forks and execs the executable