# CPSC 213

# **Introduction to Computer Systems**

### Unit 2f

## **Inter-Process Communication**

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# **Reading For Next Three Lectures**

### Textbook

- The Client Server Programming Model Web Servers
- 2nd ed: 11.1-11.5
- 1st ed: 12.1-12.5

# **IPC Basic Idea**

#### Communication for processes that don't share memory

- could be on same processor (shared physical, but not virtual) memory
- could be on different processors connected by a network
- same communication mechanism for both cases

#### Unformatted data transfer

message payload is the data to be transferred from sender to receiver
sender assembles the payload as an array of bytes -- like a file block
receiver translates byte array back into programming-language types

#### Asynchronous control transfer

- send initiate sending message payload to receiving process, but do not wait
- receive next available message, either blocking or not if no data waiting

#### Naming

- sender needs to name the receiving process
- receiver needs to name something --- options?

# **Client-Server Model**

#### server is a process that

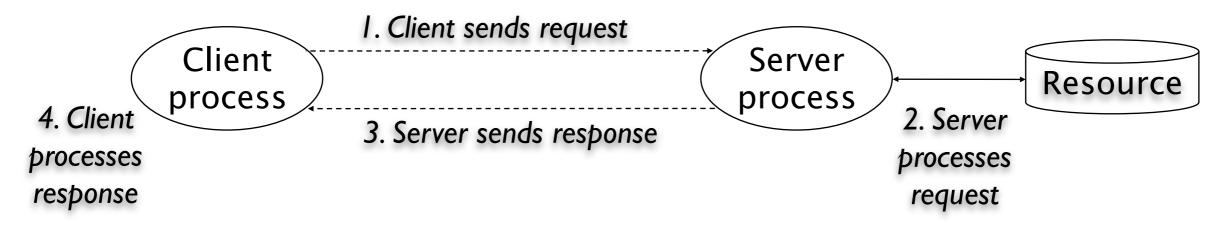
- waits to receive network messages from clients
- processes the message in some way
- sends a response message back to client
- client is a process that sends request messages to server

#### client is a process that

sends requests to server and waits for response

#### configuration

- many clients, one server
- server is often client for another server (e.g., browser, web-server, database)



Adapted from: Computer Systems: A Programmer's Perspective

# Basic communication-endpoint naming

#### Internet Protocol address (IP address)

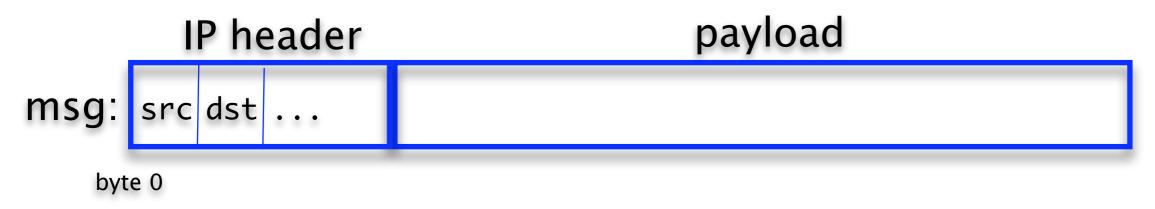
- 32-bit (IPv4) or 128-bit number
  - we write IPv4 addresses as four numbers divided by . IPv6 is 8 divided by :
- names machines nodes in an internet (there are many internets, more later)
- same-machine communication sent to 127.0.0.1 (called localhost)

### Port

- 16-bit number
- names a process, unique to a single node
- low numbers are privileged and for standard services (e.g., imap, smtp, httpd)

### Addressing a message

- destination address is IP address and port number of target process
- source address is IP address and port number of sending process
- both are included as part of the message header



# Simple example

#### sending process

allocates message buffer for payload

copies payload data into buffer

issues send

#### receiving process

issues recv to wait on port

copies payload data out of buffer and gets source address

# Determining IP address and port number

### IP Address

- usually use the IP Domain Name a hierarchical name string
  - e.g., cascade.cs.ubc.ca
- translated to IP Address by the Domain Name Service (DNS)
  - a hierarchical name server that is distributed throughout internet
  - every node is configured with the IP address of a DNS node implemented by its ISP
    - ISP is internet service provider
  - first step in communication is to contact DNS to get IP address for domain name

### port number

- some services resident on well-known ports
- you could implement your own name server for ports
- via a virtual connection using protocols like TCP

# Communication Protocols (OSI model)

### a protocol is

- a **specification** of <u>message-header formats</u> of <u>handing of messages</u>
- an implementation of the specification

### Iayering of abstraction

- several different protocols involved in sending a message
- layered into a hierarchy

#### the 7-layer OSI model (e.g., 802.11 web browsing)

<ul> <li>application</li> </ul>	HTTP	get and post etc. web-server messages
<ul> <li>presentation</li> </ul>	TCP	
<ul> <li>session</li> </ul>	TCP	
<ul> <li>transport</li> </ul>	TCP	connections, streams and reliable delivery
<ul> <li>network routing</li> </ul>	IP	routing using IP address and port #
<ul> <li>data link</li> </ul>	LLC/MAC	data framing and signalling to access airspace
<ul> <li>physical</li> </ul>	PHY	radio

# Transport protocols

### **UDP**

- send/receive datagrams
- addressing is just IP address and port # as we have seen
- best-effort transport
- but, if any router queue in network is full, message will be dropped

### TCP

- send/receive streams
- addressing using virtual connection between two hosts
- reliable, in-order delivery of stream packets
- sending rate adapts to available bandwidth
- reliability provided by software: receipt acknowledgement and retransmission

# Routing packets in the Internet

### What is the Internet

- collection of many thousands of networks bridged to each other
- backbone routers connect each of these networks
  - routing protocol is called BGP (border gateway protocol)

### Nodes are directly connected to a switch

- physical routing using, for example, ethernet or 802.11 MAC addresses
- address resolution protocol (arp)
  - broadcast protocol to resolve IP addresses on local network
- first step in sending an IP packet is to send it to your switch
- last step in receiving a packet for switch to send it to destination

### In the middle is IP and BGP routing

- each switch has set of output ports and a routing map
- map lists which IP addresses are accessible on which port
- packet is routed, step by step, eventually reaching edge router for target node
- packets have a time-to-live counter, decremented at each step, to ensure they don't travel forever in the Internet without reaching their destination
- see the route using traceroute

### finite queues at router => best-effort delivery

- if data in entering a router faster than it can leave, then packets are queued
- queues are stored in router memory and so are finite
- if queue overflows, router drops packets
- multiple paths from source to destination => out-of-order delivery
  - there are *many* paths in the network between source and destination
  - packets travel with an ISP and then between a set of ISP's
  - each ISP is connected to several others and pays money for access
  - so, an ISP might favour certain routes, because they will be cheaper
  - in any case backbone switches often have choice of which path to use
  - one factor in making the choice is relative congestion
    - if favoured route is congested, router might pick another output-port for a certain packet
  - the choice is made at each router, it is not globally co-ordinated
  - and so, packets can arrive at destination in a different order than they were sent

# Sockets

A socket is a OS abstraction for a communication endpoint

### The first step in communicating is to create a socket

- it is like opening a file (which returns a file descriptor)
- socketDescriptor = socket (domain, type, protocol)
- sd = socket ( PF\_INET, SOCK\_STREAM, 0 )
- sd = socket ( PF\_BLUETOOTH, SOCK\_STREAM, BTPROTO\_FRCOMM )
- sd = socket ( PF\_INET, SOCK\_DGRAM, 0 )

### What happens next depends on send/recv or protocol

basically, assign an network address to socket and then start communicating

### To send via UDP

create a destination address (ip address or domain name)
sendto (sd, buf, bufLen, destAddr, destAddrLen )

### To receive via UDP

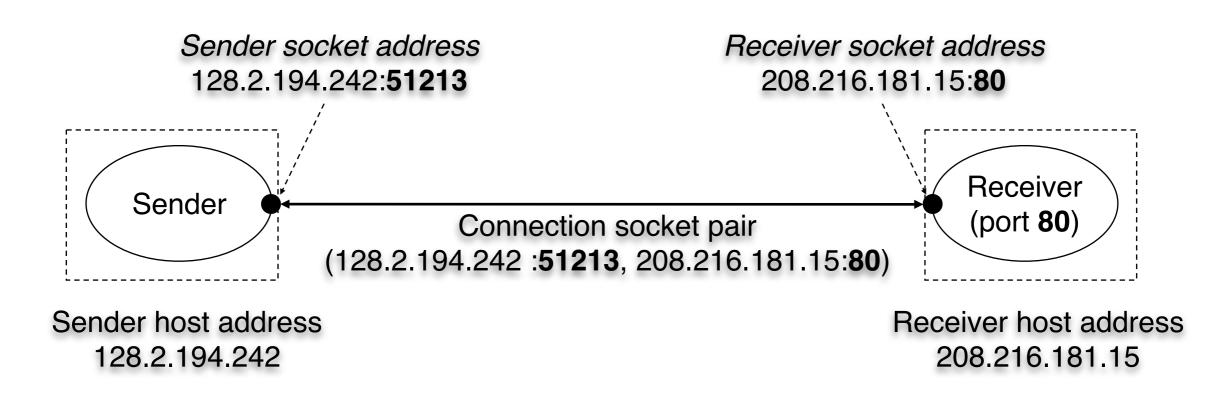
- create a receive address (port)
- bind (sd, addr, addrLen)
- recvfrom (sd, buf, bufLen, 0, srcAddr, srcAddrLen)

# **TCP Virtual Connections**

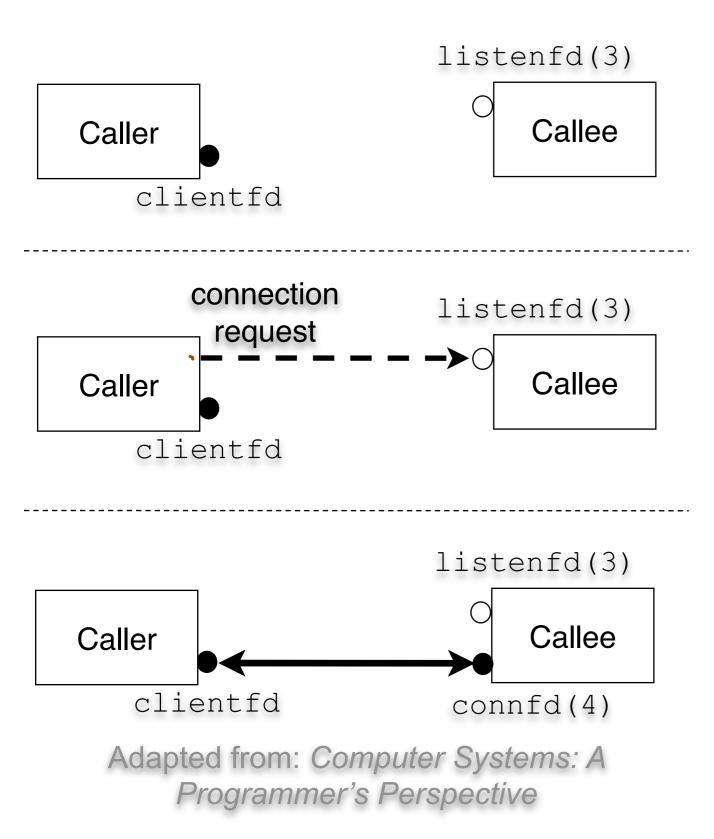
- Designed for long-term *flows* of data between a pair of endpoints
  - most traffic on internet is TCP --- otherwise Internet would not work
  - Bob Metcalfe (Ethernet inventor) eats words "catastrophic collapse" in 1995

### Basic idea

- in setup phase protocol picks send and receive port numbers for the flow
- sending application divides flow into messages
- sending OS (TCP) divides messages into packets, giving each a sequence number
- receiver sends ACK messages listing sequence numbers it is missing (roughly)
- sender retransmits these packets
- sender rate starts slow, gradually increases, slows when packet-loss rate too high



# Establishing a TCP connection

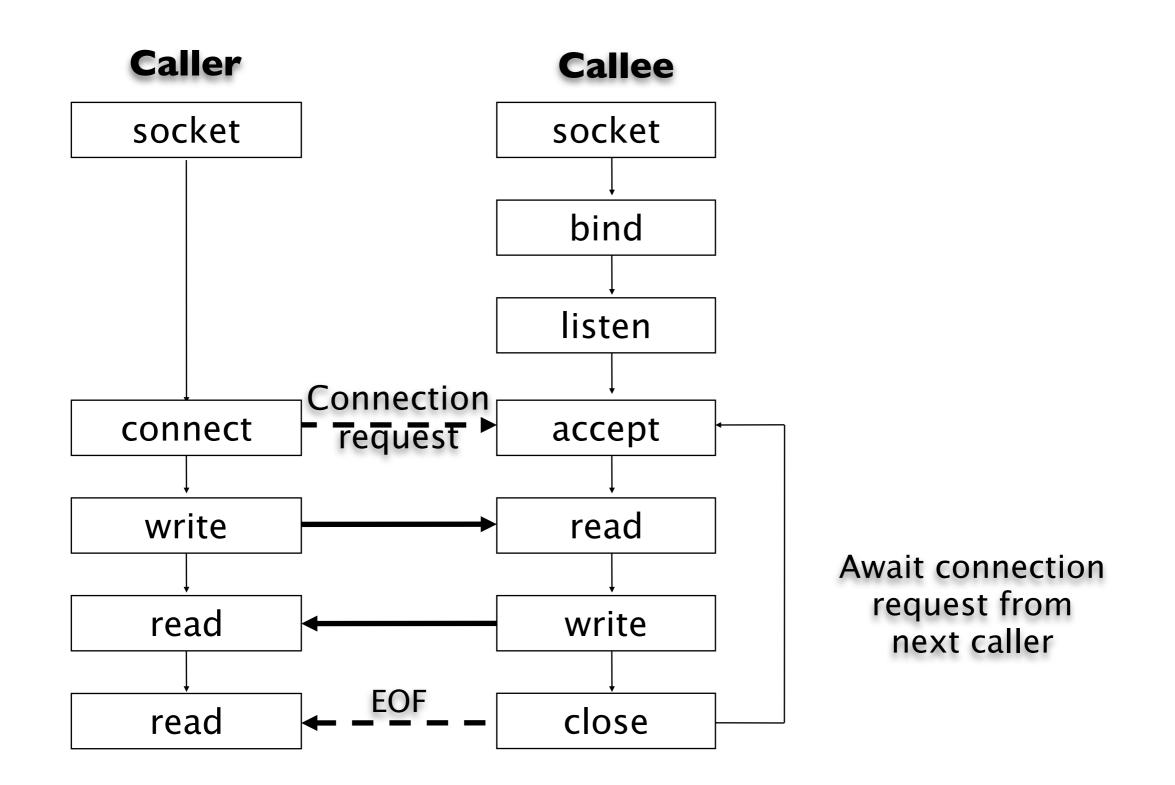


1. Callee blocks in *accept*, waiting for connection request on listening descriptor *listenfd*.

2. Caller makes connection request by calling and blocking in connect.

3. Callee returns connfd from accept. Caller returns from connect. Connection is now established between clientfd and connfd.

# Full Protocol Diagram



Adapted from: Computer Systems: A Programmer's Perspective

# **TCP Steps on Caller**

setup socket to send connection request

```
struct sockaddr_in dstAddr;
unsigned long dstIP = htonl (0xAB112090);
memset (&dstAddr, 0, sizeof (dstAddr));
dstAddr.sin_family = PF_INET;
memcpy (&dstAddr.sin_addr.s_addr, &dstIP, sizeof (dstIP));
dstAddr.sin_port = htons (7891);
```

send connection-request packet

connect (so,
 (struct sockaddr \*) &dstAddr,
 sizeof(dstAddr));

send / receive data on socket

```
send (so, buf, length, 0);
length = recv (so, buf, length, 0);
```

# TCP steps on Server

#### setup address connection-listening address

```
struct sockaddr_in conAddr;
memset(& conAddr,0,sizeof(conAddr));
conAddr.sin_family = PF_INET;
conAddr.sin_addr.s_addr = htonl(INADDR_ANY);
conAddr.sin_port = htons(7891);
bind (so, (struct sockaddr *) & conAddr,sizeof(conAddr))
```

#### setup socket to listen for connection requests

listen (so, maxNumberOfPendingRequestsQueued)

#### block waiting for connection requests to arrive

```
struct sockaddr_in caller;
int cl_len = sizeof (caller);
int callerSo;
callerSo = accept (so, (struct sockaddr *)&caller, &cl_len);
```

send/recv messages to/from caller using callerSo socket

# Summary

### Caller

1. Create a socket

- 2. Connect to callee
- 3. Transfer data
- 4. Close connection

## Callee

- 1. Create socket
- Specify contact point (binding)
- 3. Listen for calls
- 4. Accept call
- 5. Transfer data
- 6. Close connection

# A few additional details

#### purpose of bind step at server

- each machine typically has multiple network interfaces
- and so it might have multiple IP addresses
- bind picks the one to be used for this session
- bind also picks the connection-request port number (e.g., port 80)

#### finding out who called

```
struct sockaddr_in caller;
int cl_len = sizeof (caller);
int callerSo;
callerSo = accept (so, (struct sockaddr *)&caller, &cl_len);
```

unsigned long callerIP = ntohl(caller.sin\_addr.s\_addr); unsigned short = ntohs(caller.sin\_port);

#### disconnecting

close (callerSo);

# Complete Example (caller)

```
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <time.h>
int main()
{
  int fd;
  fd = socket(PF_INET, SOCK_STREAM, 0);
  struct sockaddr_in remoteAddr;
  unsigned long remoteIP =
                         htonl(0x7F000001);
  memset(&remoteAddr, 0,
              sizeof(remoteAddr));
  remoteAddr.sin_family = PF_INET;
  memcpy(&remoteAddr.sin_addr.s_addr,
                &remoteIP, sizeof(remoteIP));
  remoteAddr.sin_port = htons(7891);
```

```
if (connect(fd, (struct sockaddr *) &remoteAddr,
  sizeof(remoteAddr)) < 0) {</pre>
    perror("Connection failed");
  } else {
    char *msg = "Hi there\n";
    time_t ltime;
    char buff[256];
    time(&ltime);
    write(fd, &ltime, sizeof(ltime));
    write(fd, msg, 10);
    int amt;
    amt = read(fd, buff, 256);
    while(amt > 0) {
      printf("Buffer %s", buff);
      amt = read(fd, buff, 256);
    }
 }
}
```

# Complete Example (server)

```
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <netdb.h>
#include <time.h>
int main()
{
  int fd;
  fd = socket(PF_INET, SOCK_STREAM, 0);
  struct sockaddr_in callerAddr;
  memset(&callerAddr, 0, sizeof(callerAddr));
  callerAddr.sin_family = PF_INET;
  callerAddr.sin_addr.s_addr = htonl(INADDR_ANY);
  callerAddr.sin_port = htons(7891);
  bind(fd, (struct sockaddr *) &callerAddr,
      sizeof(callerAddr));
  listen(fd, 4);
  struct sockaddr_in caller;
  while(1) {
    int cl_len = sizeof(caller);
    int callerFD;
    callerFD = accept(fd,
  (struct sockaddr *)&caller,
  &cl_len);
```

```
char buff[256];
 int amt;
  time_t rtime;
  recv(callerFD, &rtime, sizeof(time_t), 0);
  amt = recv(callerFD, buff, 256, 0);
  printf("%s", buff);
  struct hostent *hp;
 hp = gethostbyaddr((char *)
         &caller.sin_addr.s_addr,
          sizeof(long), PF_INET);
  amt = snprintf(buff,256,
       "Connection from %s (%x) %x at %s",
       hp->h_name,
       ntohl(caller.sin_addr.s_addr),
       (long) ntohs(caller.sin_port),
       ctime(&rtime));
  send(callerFD, buff, amt + 1, 0); sleep(10);
  send(callerFD, "Bye\n", 5, 0);
  close(callerFD);
}
```

}

# **BSD Socket API Summary**

- socket() creates the socket
- connect() initiate a connection
- bind() indicates the IP address to use
- Isten() marks the socket to receive connections
- read()/recv() reads data
- write()/send() writes data
- close()
- accept()

- shuts down the connection
- waits for incoming connection

# Other useful functions

- inet\_aton() string to network address
- inet\_ntoa() network address to string
- gethostbyname() lookup host by IP domain name (get hostent)
- gethostbyaddr() lookup host by IP address

# A naive web server

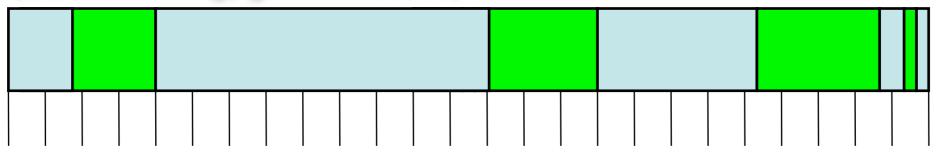
while (1) {

accept connection perform http request close connection

### **Request Timeline:**

}

(blue is waiting, green is active)



- 1.wait for request
- 2.process request , read from file
- 3.wait for file read to complete

### What is wrong?

4.may repeat 2 & 3 several times

- 5.prepare reply and send
- 6.goto step 1