# Computer Networks

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Lecture 1

## Introduction - Administrative

#### > Weekly:

- Lectures
- Labs Each lab we practice things (programming and networking simulations).
- You do not have individual scheduled assignments but you need to be able to solve any problem from the proposed set

#### > Final grade (average of):

- After each lab session (approx. 6 weeks) there will be a test counting towards your final grade
- Final written examination / One Moodle test towards the end of semester

### Prerequisites

- C/C++ system programming (Unix and Windows)
- Operating systems
- Python, PHP (interpreter)

# Bibliography

- J. Kurose, K. Ross, Computer Networking: A Top Down Approach, Addison-Wesley, rev2,3,4 2002-2007.
- A.S. Tanenbaum Computer Networks 4th ed., Prentice Hall, 2003
- Douglas E. Comer, Internetworking with TCP/IP
   Vol 1- Principles, Protocols, and Architecture
   Vol 3- Client-Server Programming and Applications
- G.R.Wright, R. Stevens, TCP/IP Illustrated vol 1,2, Addison Wesley.
- Matt Naugle, Illustrated TCP/IP A Graphic Guide to protocol suite, John Willey & Sons, 1999.
- W. Richard Stevens, Bill Fenner, Andrew M. Rudoff, UNIX® Network Programming Volume 1, Third Edition: The Sockets Networking **ĀPI**

# **Course Information**

http://www.cs.ubbcluj.ro/~dadi/compnet

# Required (?!) Tools/Materials

- Windows 32/64 +Linux Development Env!
- VMware Player/ Virtual Box, etc
  - Install Linux / Windows!
  - Integration Tools
  - Development Environment Or Vi ?!?
- Set networking as <u>bridged</u> on VM!
  You will thank me later :P

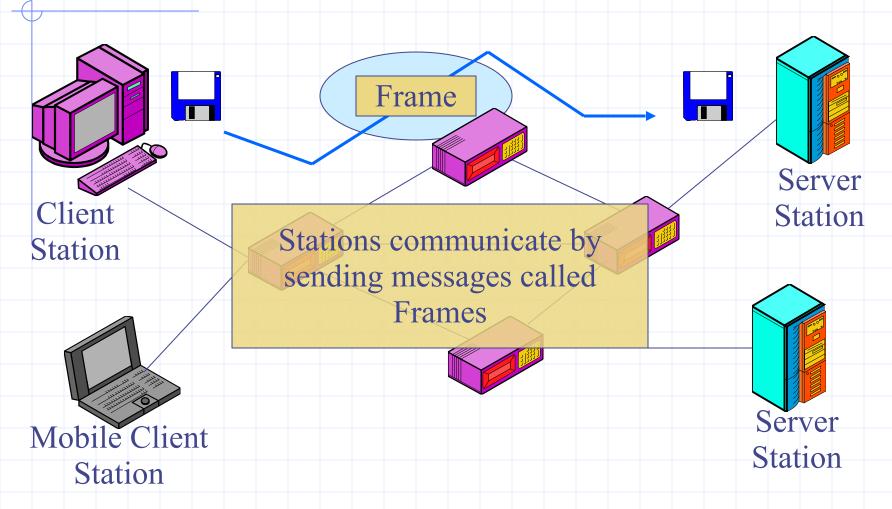
## Syllabus – What are we going to study?

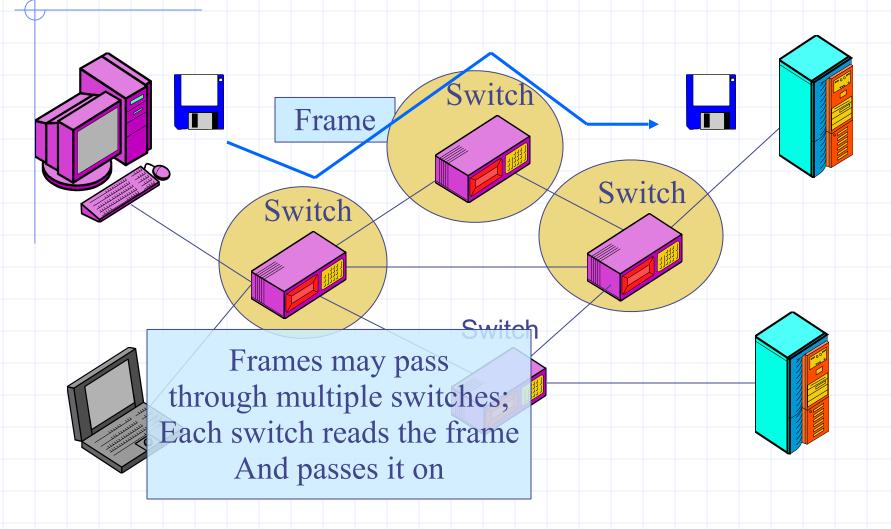
- TCP/IP Programming
  - 1. TCP sockets
  - 2. UDP sockets
- 2. Communication Protocols and Hierarchies
- 3. The OSI and TCP/IP layered architecture
- 4. The IP Protocol
  - 1. IP Addressing schemas
  - 2. Helper protocols: DHCP, ARP, DNS
- TCP and UDP
- 6. Routing

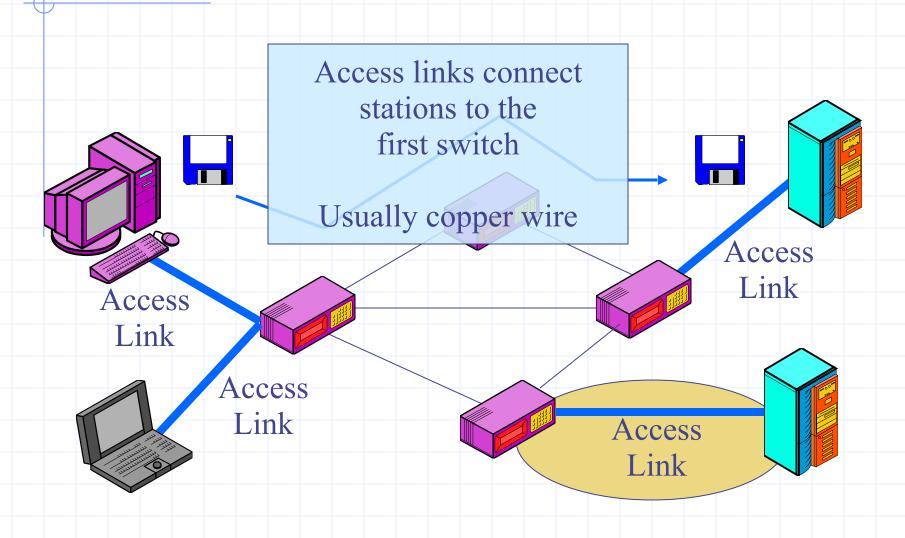
A collection of computers (PCs, Workstations) and other devices interconnected.

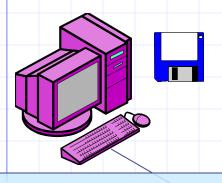
## **Components:**

- Hosts (computers)
- Links (coaxial cable, twisted pair, optical fiber, radio, satellite)
- Switches/routers (intermediate systems)





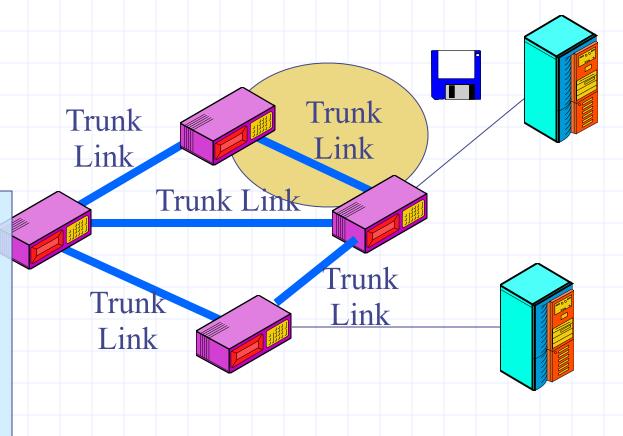




Trunk links connect switches

Higher capacity than access links

Often optical fiber



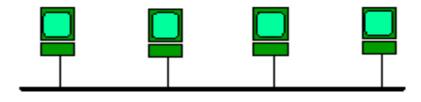
## Classifications

- Types of links
  - Direct links
  - Bus type links
- Type of transmission
  - Circuit switched networks
  - Packet switched networks
  - Frame Relay
  - Asynchronous Transfer Mode (ATM)

# Types of communication

1. Types of links (connectivity)



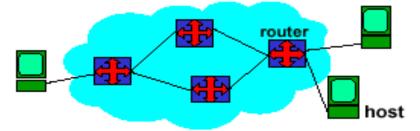


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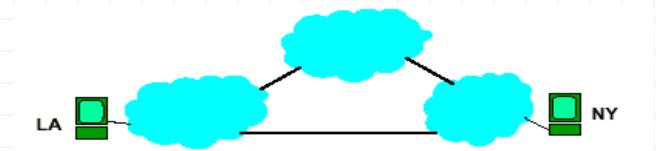
# Types of Communication

### **Switched Networks**

Circuit - switched network: public telephone network

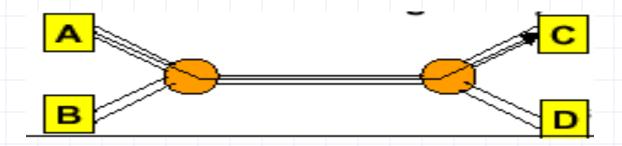


 Packet switched network: Internet (collection of networks)



# Circuit-Switching

- Set up a connection path (circuit) between the source and the destination (permanent for the lifetime of the connection)
- network resources (e.g., bandwidth) divided into "pieces".
  Pieces allocated to calls
- Resource piece idle if not used by owning call (no sharing)
- All bytes follow the same dedicated path
- ➤ While A talks to C, B cannot talk to D on the same line.
- Piece Division ? TimeSlot vs frequency division



## Packet Switching

- each end-end data stream divided into packets
- user A, B packets share network resources
- each packet uses full link bandwidth
- > resources used as needed

Bandwidth division into 'pieces"

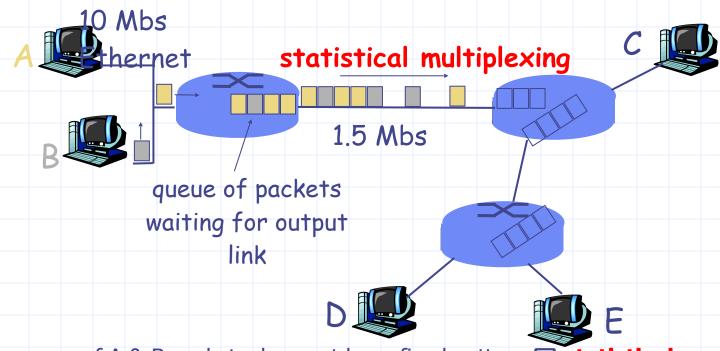
Dedicated allocation

Resource reservation

#### resource contention:

- aggregate resource demand can exceed amount available
- congestion: packetsqueue, wait for link use
- store and forward: packets move one hop at a time
  - transmit over link
  - wait turn at nextlink

### Packet Switching: Statistical Multiplexing

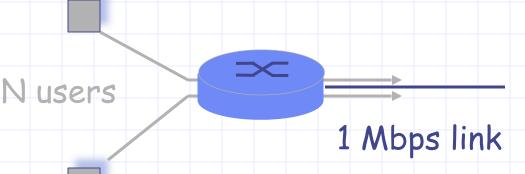


- Sequence of A & B packets does not have fixed pattern ? statistical multiplexing.
- > Efficient use of resources: Nobody reserves a lane on a freeway!
- Can accommodate bursty traffic (as opposed to circuit-switching where transmission is at constant rate).

## Packet switching versus circuit switching

Packet switching allows more users to use network!

- ≥ 1 Mbit link
- > each user:
  - 100 kbps when "active"
  - active 10% of time
- circuit-switching:
  - 10 users
- packet switching:
  - with 35 users, probability10 active less than.0004

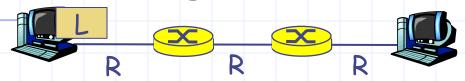


### Packet switching versus circuit switching

Is packet switching a "slam dunk winner?"

- Great for bursty data
  - resource sharing
  - simpler, no call setup
- Excessive congestion: packet delay and loss
  - protocols needed for reliable data transfer, congestion control
- Q: How to provide circuit-like behavior?
  - bandwidth guarantees needed for audio/video apps
  - still an unsolved problem (chapter 6)

## Packet-switching: store-and-forward

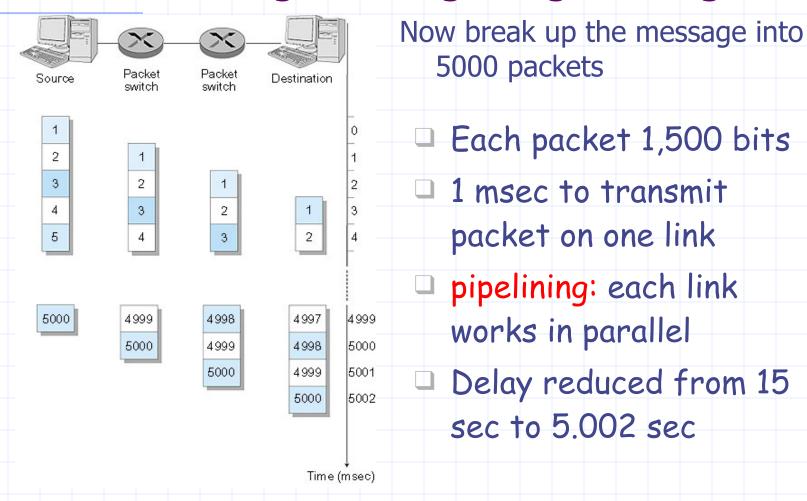


- Takes L/R seconds to transmit (push out) packet of L bits on to link or R bps
- Entire packet must arrive at router before it can be transmitted on next link: store and forward
- $\rightarrow$  delay = 3L/R

#### Example:

- ightharpoonup L = 7.5 Mbits
- > R = 1.5 Mbps
- $\rightarrow$  delay = 15 sec

## Packet Switching: Message Segmenting



# App-layer protocol defines

- Types of messages exchanged, e.g., request & response messages
- Syntax of message types: what fields in messages & how fields are delineated
- Semantics of the fields, i.e., meaning of information in fields
- Rules for when and how processes send & respond to messages

#### Public-domain protocols:

- defined in RFCs
- allows for interoperability
- eg, HTTP, SMTP

#### Proprietary protocols:

eg, KaZaA

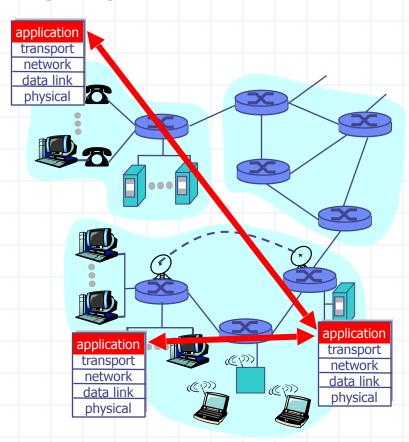
### Applications and application-layer protocols

# Application: communicating, distributed processes

- e.g., e-mail, Web, P2P file sharing, instant messaging
- running in end systems (hosts)
- exchange messages to implement application

#### Application-layer protocols

- one "piece" of an app
- define messages exchanged by apps and actions taken
- use communication services provided by lower layer protocols (TCP, UDP)



Client-server paradigm

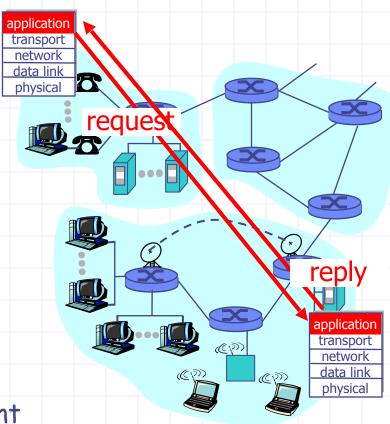
Typical network app has two pieces: client and server

#### Client:

- initiates contact with server ("speaks first")
- > typically requests service from server,
- Web: client implemented in browser; e-mail: in mail reader

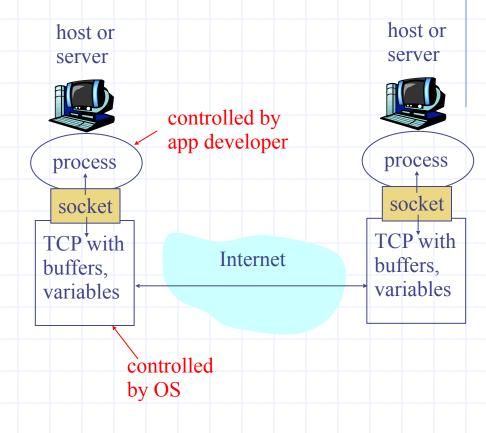
#### Server:

- > provides requested service to client
- > e.g., Web server sends requested Web page, mail server delivers e-mail



### Processes communicating across network

- process sends/receives messages to/from its socket
- socket analogous to door
  - sending process shoves message out door
  - sending process assumes transport infrastructure on other side of door which brings message to socket at receiving process



> API: (1) choice of transport protocol; (2) ability to fix a few parameters (lots more on this later)

### Addressing processes:

- For a process to receive messages, it must have an identifier
- Every host has a unique32-bit IP address
- Q: does the IP address of the host on which the process runs suffice for identifying the process?
- Answer: No, many processes can be running on same host

- Identifier includes both the IP address and port numbers associated with the process on the host.
- Example port numbers:
  - > HTTP server: 80
  - Mail server: 25
- ➤ More on this later

### What transport service does an app need?

#### **Data loss**

- some apps (e.g., audio) can tolerate some loss
- other apps (e.g., file transfer, telnet,ssh) require
   100% reliable data transfer

#### **Timing**

some apps (e.g., Internet telephony, interactive games) require low delay to be "effective"

#### Bandwidth

- > some apps (e.g., multimedia) require minimum amount of bandwidth to be "effective"
- other apps ("elastic apps") make use of whatever bandwidth they get

### Transport service requirements of common apps

Application	Data loss	Bandwidth	Time Sensitive
file transfer	no loss	elastic	no
e-mail	no loss	elastic	no
Web documents	no loss	elastic	no
real-time audio/video	loss-tolerant	audio: 5kbps-1Mbps video:10kbps-5Mbps	<del>yes, 100's m</del> sec
stored audio/video interactive games instant messaging	loss-tolerant loss-tolerant no loss	same as above few kbps up elastic	yes, few secs yes, 100's msec yes and no

### Internet transport protocols services

#### TCP service:

- connection-oriented: setup required between client and server processes
- reliable transport between sending and receiving process
- flow control: sender won't overwhelm receiver
- congestion control: throttle sender when network overloaded
- does not provide: timing, minimum bandwidth guarantees

#### **UDP** service:

- unreliable data transfer between sending and receiving process
- does not provide:
   connection setup, reliability,
   flow control, congestion
   control, timing, or
   bandwidth guarantee

Q: why bother? Why is there a UDP?

### Internet apps: application, transport protocols

Application	Application layer protocol	Underlying transport protocol
e-mail	SMTP [RFC 2821]	TCP
remote terminal access Web	Telnet [RFC 854] HTTP [RFC 2616]	TCP
file transfer streaming multimedia	FTP [RFC 959] proprietary	TCP or UDP
Internet telephony	(e.g. RealNetworks)  proprietary (e.g., Dialpad)	typically UDP

# Network programming

- Programmer does not need to understand the hardware part of network technologies.
- Network facilities accessed through an Application Program Interface - API

# Socket programming

Goal: learn how to build client/server application that communicate using sockets

#### Socket API

- introduced in BSD4.1 UNIX, 1981
- explicitly created, used, released by apps
- client/server paradigm
- two types of transport service via socket API:
  - unreliable datagram
  - reliable, byte streamoriented

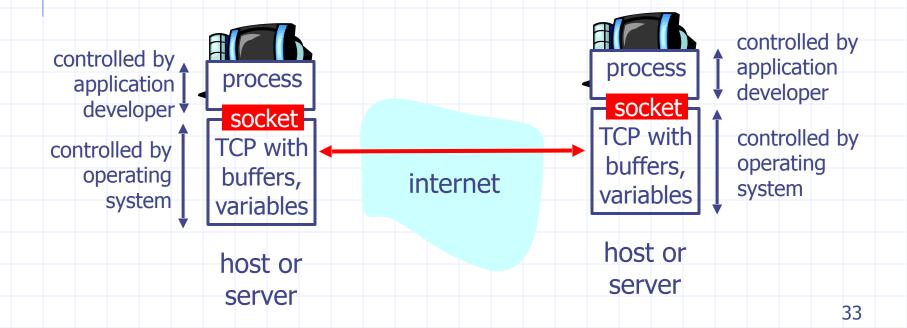
#### socket

a host-local,
application-created,
OS-controlled interface (a
"door") into which
application process can
both send and
receive messages to/from
another application process

## Socket-programming using TCP

Socket: a door between application process and end-end-transport protocol (UCP or TCP)

TCP service: reliable transfer of bytes from one process to another



## Socket programming with TCP

#### Client must contact server

- server process must first be running
- server must have created socket (door) that welcomes client's contact

#### Client contacts server by:

- creating client-local TCP socket
- specifying IP address, port number of server process
- When client creates socket: client TCP establishes connection to server TCP

- When contacted by client, server TCP creates new socket for server process to communicate with client
  - allows server to talk with multiple clients
  - source port numbers used to distinguish clients (more in Chap 3)

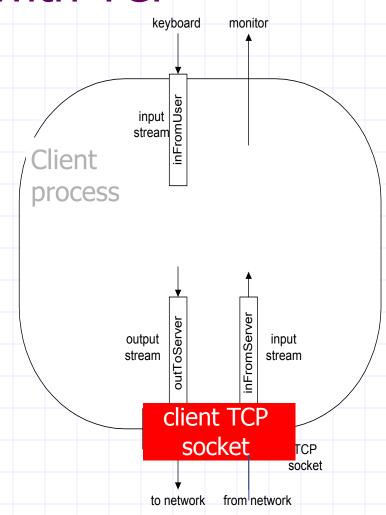
#### application viewpoint

TCP provides reliable, in-order transfer of bytes ("pipe") between client and server

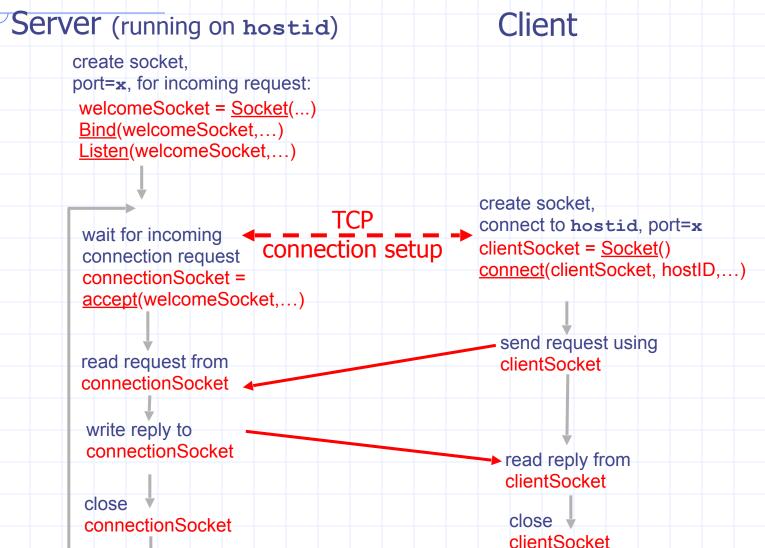
Socket programming with TCP

#### Example client-server app:

- 1) client reads line from standard
   input (inFromUser stream),
   sends to server via socket
   (outToServer stream)
- 2) server reads line from socket
- 3) server sends back to client (echo) the same data
- 4) client reads from socket and prints line (inFromServer stream)



### Client/server socket interaction: TCP



## Connection oriented-API

- The BSD socket library
  - Socket
  - Bind
  - Listen, Accept
  - Connect
  - Read, Write, Recv, Send
  - Close, Shutdown
- Where do we get info on these ?
  - man, msdn

# Socket Example

#### Server.c

```
#include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#include <netdb.h>
#include <stdio.h>
#include <stdio.h>
#include <unistd.h> /* close */
#define SERVER_PORT 1500
```

```
int main (int argc, char *argv[]) {
 int sd, newSd, cliLen;
 struct sockaddr_in cliAddr, servAddr;
 char line[MAX_MSG];
 int len;
 sd = socket(AF_INET, SOCK_STREAM, 0);
 if(sd<0) {
  perror("cannot open socket ");
  return ERROR;
 /* bind server port */
 servAddr.sin_family = AF_INET;
 servAddr.sin_addr.s_addr = htonI(INADDR_ANY);
 servAddr.sin_port = htons(SERVER_PORT);
```

```
if (<u>bind</u>(sd, (struct sockaddr *)
   &servAddr, sizeof(servAddr))<0) {
  perror("cannot bind port ");
 return ERROR;
<u>listen</u>(sd,5);
while(1) {
  printf("%s: waiting for data on port
   TCP %u\n",argv[0],SERVER_PORT);
 cliLen = sizeof(cliAddr);
 newSd = <u>accept</u>(sd, (struct sockaddr
   *) &cliAddr, &cliLen);
 if(newSd<0) {
   perror("cannot accept connection ");
   return ERROR;
  } // end if
```

```
/* init line */
  memset(line,0,MAX_MSG);
  /* receive segments */
 if ( (len=read(newSd,line,MAX_MSG))> 0) {
   printf("%s: received from %s:TCP%d:
   %s\n", argv[0],
      inet_ntoa(cliAddr.sin_addr),
       ntohs(cliAddr.sin_port), line);
    write(newSd,line,len);
  } else
    printf("Error receiving data\n");
  close(newSd);
 } //end if
} //end while
```

#### **CLIENT.C**

```
include <sys/types.h>
#include <sys/socket.h>
#include <netinet/in.h>
#include <arpa/inet.h>
#include <netdb.h>
#include <stdio.h>
#include <unistd.h> /* close */
#define SERVER PORT 1500
#define MAX MSG 100
int main (int argc, char *argv[]) {
 int sd, rc, i;
 struct sockaddr_in servAddr;
 struct hostent *h;
 char msg[300];
```

```
if(argc < 3) {
 printf("usage: %s <server> <text>\n",argv[0]);
 exit(1);
h = gethostbyname(argv[1]);
if (h==NULL) {
 printf("%s: unknown host
  '%s'\n",argv[0],argv[1]);
 exit(1);
servAddr.sin_family = h->h_addrtype;
memcpy((char *) &servAddr.sin_addr.s_addr,
  h->h_addr_list[0], h->h_length);
servAddr.sin_port = htons(SERVER_PORT);
```

```
/* create socket */
 sd = socket(AF_INET, SOCK_STREAM, 0);
 if(sd<0) {
  perror("cannot open socket ");
  exit(1);
 /* connect to server */
 rc = connect(sd, (struct sockaddr *) &servAddr, sizeof(servAddr));
 if(rc<0) {
  perror("cannot connect ");
  exit(1);
 write(rc, argv[1],strlen(argv[1]+1) );
 read(rc, msg, 300);
 printf("Received back: %s\n", msg);
 close(rc);
 return 0;
```