Communication Networks (0368-3030) / Spring 2011 The Blavatnik School of Computer Science, Tel-Aviv University

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Homework

- · 3 practical assignments
 - "hands-on" network programming
 - C / C++
- 4-5 theoretical assignments
- will probably include some guided-reading bonus points
- Guided-reading is considered part of the material for the final exam
- Moodle forum for HW related questions

Requirements & Grading

- Final Exam 60%
- Practical HW assignments 20%
- Theoretical HW assignments 20%
- · Submission of all the assignments is mandatory
- HW may be submitted in pairs
- There will be a closed-books final exam
 You may bring 4 pages (i.e. 2 two-sided sheets) with you to the exam

Textbooks & Online Material

- Course website:
- http://www.cs.tau.ac.il/~allonwag/comnet2013A/index.html • Main textbook:
- Computer Networking: A Top-down Approach, by J. F. Kurose and K. W. Ross (3rd edition or later).
- · Other references:
- Computer Networks, by A. S. Tanenbaum (4th edition or later).
- Computer Networks: A Systems Approach, by L. L. Peterson and B. S. Davie (3rd edition or later).
- An Engineering Approach to Computer Networking, by S. Keshav.
- Wikipedia, and lots of online material

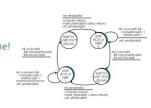
Why study computer networks?

- An interface between theory (algorithms, mathematics) and practice
- Understanding the design principles of a truly complex system
- Industry-relevant knowledge
- Fun!
- · Challenges in teaching computer networks
- Students' feedback

Introduction

Protocols

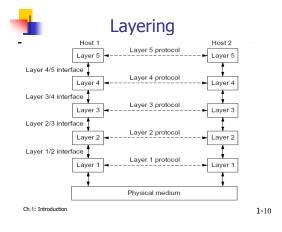
- A protocol defines:
- Format (Syntax)
- Conversation logic
- → Finite state machine!
- Open/ proprietary



Networking is a complex task

Solution: modularity

- Layering
- Transparency
- · Each layer is dependent only on the interfaces defined by the layers above and below it
- · Each layer "talks" only to its equivalent on the remote side
- · Each layer is implemented by a protocol



Layering Models

- OSI Reference Model
 - 7 layers
 - Defined by ISO (International Standards Organization) Widely used as a reference model, but seldom
- implemented

• TCP/IP Reference Model

- 5 layers
- Protocols came first, the model is actually a description of their workings.
- The TCP/IP suite is the backbone of today's Internet.

Overview of the 5-layers model

- Physical layer
 - Transmits raw bits over a communication channel
- Data link layer
- Control layer over the physical layer • Framing
- Network layer
 - Delivers packets from source to destination across the network
- Routing vs. Forwarding
- In TCP/IP: IP is the forwarding protocol

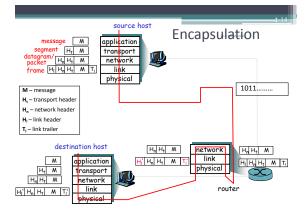
application

- transport
- network
- data link
- physical



- First end-to-end layer!
- In TCP/IP:
- TCP: reliable, connection-oriented
 UDP: unreliable, connectionless
- Application layer
 - A protocol (sometimes a protocol stack) to implement the desired application service.
 - Examples:
 Mail: SMTP, POP3, IMAP
 - Remote control: Telnet
 - File transfer and sharing: FTP, Bittorrent
 - Instant messaging: XMPP (Jabber)

application
transport
network
data link
physical



HW Objective: Write a network application

- Design an application protocol
 - Syntax
 - Semantics
 - Conversation logic
- Implement via socket programming
- An interface to the OS's transport layer

Socket Programming – Part I

Recommended References: Beej's Guide to Network Programming http://beej.us/guide/bgnet/ Unix Network Programming \ W. Richard Steven

Slides for this topic, as well as other topics along the course, are partly based on the work of previous teaching assistants to this course: Hillel Avni, Yahav Nussbaum, David Raz, Motti Sorani, Alex Kesselman.

IP Address / Domain Names

- "Uniquely" identifies a "host" on the network • Not really, we'll get to that later in the course
- A 32-bit number
 - For convenience represented as 4 numbers in the range 0-255
 - e.g. 132.67.192.133
- Domain names
 - 132.67.192.133 = nova.cs.tau.ac.il

Port

- A 16-bit number (i.e., 0-65535)
- Identifies a service on the host
 - Again, not quite, we'll get to that later, blah-blah.
 - For instance: HTTP = 80, SMTP = 25, Telnet = 23
- A socket is a combination of IP + port
 132.67.192.133:80

Port (cont.)

- The server listens on a certain port
- The client randomly chooses a port to which the server answers
- For instance
 94.127.73.5 : 1902 ↔132.67.192.133 : 80

Relevant Headers

- #include <sys/socket.h>
 Sockets
- #include <netinet/in.h>

 Internet addresses
- #include <arpa/inet.h>
 - Working with Internet addresses
- #include <netdb.h>
- Domain Name Service (DNS)
- #include <errno.h>
 - Working with errno to report errors

Address Representation

```
struct sockaddr {
   u_short sa_family;
   char sa_data[14];
};
```

sa_family

- specifies which address family is being used
- determines how the remaining 14 bytes are used

Address Representation – Internet Specific

```
struct sockaddr_in {
   short sin_family; /* = AF_INET */
   u_short sin_port;
   struct in_addr sin_addr;
   char sin_zero[8]; /* unused */
};
struct in_addr {
   uint32_t s_addr;
}
• Except for sin_family, all contents are in network order
```

Big Endian / Little Endian

- Memory representation of multi-byte numbers:
 - \sim 2882400018₁₀ = ABCDEF12₁₆
 - Big Endian: 0xAB CD EF 12
 - Little Endian: 0x 12 EF CD AB
- Hosts on the web use both orders
- On the network all use big endian (= network order).
- Numbers used for port number, IP etc. should thus be converted
 - htonl () / ntohl() / htons() / ntohs()

Reliable vs. Unreliable Sockets

SOCK_STREAM	SOCK_DGRAM
reliable transport	unreliable transport
connection-oriented	connectionless
keeps state	stateless
more resources needed	lightweight
ТСР	UDP

Session overview

• We will start with reliable transport (TCP)

Client	ТСР	Server
		socket()
		bind()
socket()		listen()
connect()	$\leftarrow \text{session setup} \rightarrow$	accept()
send()	data transfer \rightarrow	recv()
recv()	← data transfer	send()
close()	\leftarrow terminate session \rightarrow	close()

Socket Creation – socket()

- int socket(int domain, int type, int protocol);
- domain: PF_INET for IPv4
- type: for our purposes either SOCK_STREAM or SOCK_DGRAM
- protocol: can be set to 0 (default protocol)
- Returns the new socket descriptor to be used in subsequent calls, or -1 on error (and errno is set accordingly).
- Don't forget to close the socket when you're done with it

Bind socket to IP and port - bind()

- int bind(int sockfd, const struct sockaddr *my_addr, socklen_t addrlen);
- sockfd : socket descriptor
- my_addr: address to associate with the socket
 The IP portion often set to INADDR_ANY which means "local host"
- addrlen: set to sizeof(my_addr)
- Returns 0 on success, or -1 on error (and errno is set accordingly).

Wait for an incoming call - listen()

- int listen(int sockfd, int backlog);
- sockfd : socket descriptor
- backlog: number of pending clients allowed, before starting to refuse connections.
- Returns 0 on success, or -1 on error (and errno is set accordingly).

Accept an incoming connection – accept()

- int accept(int sockfd, struct sockaddr *addr, socklen_t *addrlen);
- sockfd : socket descriptor
- addr: filled in with the address of the site that's connecting to you.
- addrlen: filled in with the sizeof() the structure returned in the *addr* parameter
- Returns the newly connected socket descriptor, or -1 on error, with *errno* set appropriately.
- Don't forget to close the returned socket when you're done with it

Server-side example

sock = socket(PF_INET, SOCK_STREAM, 0);

myaddr.sin family = AF INET; myaddr.sin_port = hton5(80); myaddr.sin_addr = hton1(INADDR_ANY);

bind(sock, &myaddr, sizeof(myaddr));

listen(sock, 5);

sin_size = sizeof(struct sockaddr_in); new_sock = accept(sock, (struct sockaddr*) &their_addr, &sin_size);

• In real-life code, don't forget to check for errors

Session overview

• Reliable transport (TCP)

Client	тср	Server
		socket()
		bind()
socket()		listen()
connect()	\leftarrow session setup \rightarrow	accept()
send()	data transfer \rightarrow	recv()
recv()	← data transfer	send()
close()	←terminate session→	close()

Connect to a listening socket - connect()

- int connect(int sockfd, const struct sockaddr *serv_addr, socklen_t addrlen);
- sockfd : socket descriptor
- serv_addr: the address you're connecting to.
- addrlen: filled with sizeof(serv_addr)
- Returns 0 on success, or -1 on error (and errno is set accordingly).
- Most of the times, no bind() is required on the client side:
 If bind() wasn't called, the local IP address and a random high port are used.

Client-side example

sock = socket(PF_INET, SOCK_STREAM, 0);

dest_addr.sin_family = AF_INET; dest_addr.sin_port = htons(80); dest_addr.sin_addr = htonl(0x8443FC64);

- connect(sock, (struct sockaddr*)
 &dest_addr, sizeof(struct sockaddr));
- In real-life, the server's IP is not hard-coded
- In real-life code, don't forget to check for errors

Session overview

- Once the session is initiated, both parties are equal:
- Both can send and receive data
- Both can decide it's time to close the connection
- As long as the listening socket is open, it can accept new incoming clients
 - by calling accept()

Active	Passive
socket()	socket()
	bind()
	listen()
connect()	accept()
Conne	ected
close()	close()
	accept()

Closing a connection – close()

- int close(int sockfd);
- sockfd : socket descriptor
- returns 0 on success, or -1 on error (and errno is set accordingly)
- After we close a socket:
- If the remote side calls recv(), it will return 0.
- If the remote side calls send(), it will receive a signal SIGPIPE and send() will return -1 and errno will be set to EPIPE.
- shutdown() can be used to close only one side of the session
 Rarely used
 - Refer to the man pages

Session overview

• Unreliable transport (UDP)

Client	UDP	Server
		socket()
socket()		bind()
sendto()	data transfer $ ightarrow$	recvfrom()
recvfrom()	← data transfer	sendto()
close()		close()

Sending data (TCP + UDP)

- * TCP: ssize_t send(int socket, const void *buffer, size_t length, int flags);
- UDP: ssize_t sendto(int socket, const void *buffer, size_t length, int flags, const struct sockaddr *dest_addr, socklen_t dest len);
- buffer, length: buffer of the data to send, and number of bytes to send from it.
- flags: send options. Refer to the man pages. Use 0 for "no options".
- In unconnected sockets (UDP) you specify the destination in each sendto().

Partial send

- send() and sendto() return the number of bytes actually sent, or -1 on error (and errno is set accordingly).
- The number of bytes actually sent might be less than the number you asked it to send.

A code considering that

(Use it for TCP. For UDP it makes less sense – we will discuss later)

- 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;</pre>
- *len = total; // return number actually sent here return n == -1 ? -1:0; //-1 on failure, 0 on success }

Source: Beej's Guide to Network Programming

Receiving data (TCP + UDP)

- TCP: ssize_t recv(int socket, void *buffer, size_t length, int flags);
- * UDP: ssize_t recvfrom(int socket, void *buffer, size_t length, int flags, struct sockaddr *from_addr, socklen_t from_len);
- buffer, length: allocated space for the received data, and its size (= max data received by this call)
- flags: receive options. Refer to the man pages. Use 0 for "no options".

Receiving data (TCP + UDP) (cont.)

- recv() and recvfrom() return the number of bytes received, or -1 if an error occurred (and errno is set accordingly).
- In TCP sockets, 0 is returned if the remote host has closed its connection.
 This is often used to determine if the remote side has
- closed the connection.
 In unconnected sockets (UDP) from_addr will hold upon return the source address of the received message.
- from_len should be initialized before the call to sizeof(from_addr). It is modified on return to indicate the actual size of the address stored in from_addr.

Translating a host name to an IP address

- struct hostent *gethostbyname(const char *name);
- deprecated
 int getaddrinfo(const char *hostname, const char *servname, const struct addrinfo *hints, struct addrinfo **res);
- addrinfo *hints, struct addrinfo **res);
 Supports many options and thus seems complex, but basic
 use is simple.
 - Refer to Beej's guide for more info and for a simple example of its use: http://beej.us/guide/bgnet/output/html/multipage/getaddrinfo
 - man.html Don't forget to use *freeddrinfo*() to release memory when
- Don't forget to use *freeaddrinfo*() to release memory when you're done with *getaddrinfo*'s result.

Other Useful Functions

- inet_ntop(), inet_pton()
- Convert IP addresses to human-readable text and back
 getpeername()
- Return address info about the remote side of the connection.
- Used after calling accept() (server) or connect() (client)
- gethostname()
- returns the standard host name for the current processor

What do we send?

Tips for defining a protocol

Binary protocols

- Uniform endianity for numbers
- String representation:
 Bad: decide on maximal length
- hello = 0x 68 65 6C 6C 6F 00 00 00 00 • Better: use a length field
- hello = 0x 05 00 68 65 6C 6C 6F (note that the integer is in little endian)
- Length field can also be applied to fields of variable length (e.g., options)

An example:

- A DNS response for the query www.icann.org: 91 73 81 80 00 01 00 01 00 00
- 00 00 03 77 77 77 05 69 63 61 6e 6e 03 6f 72 67 00 00 01 00 01 c0 0c 00 01 00 01 00 00 02 58 00 04 c0 00 20 07
- For instance, bytes 0-1 are transaction ID, bytes 2-3 hold various flags.
- Text view:
 .s....www
 - .icann.org...... X.... .

Textual Protocols – An example

HTTP request for the page http://www.ietf.org/rfc/rfc3514.txt

GET /rfc/rfc3514.txt HTTP/1.1 Host: www.ietf.org

- riost: www.lett.org Accept: text/ihrml,application/xhtml+xml,a pplication/xml;q=0.9,*/*;q=0.8 Accept-Language: en-us,en;q=0.5 Accept-Encoding: gzlp,deflate Accept-Charset: ISO-8859-1,utf-
- 8;q=0.7,*;q=0.7 Keep-Alive: 115

Connection: keep-alive

HTTP/1.1 200 OK Date: Sun, 13 Feb 2011 14:32:45 GMT Last-Modified: Fri, 28 Mar 2003 18:36:14 GMT Content-Length: gzip Content-Length: 4486 Keep-Alive: timeout=15, max=100 Connection: Keep-Alive Content-Type: text/plain

The response:

Know the difference between TCP and UDP

ТСР

- Reliable
 Transfers a stream of data
 send() and recv() do not
 - necessarily match message boundaries!
 Can receive multiple messages
 - can receive multiple messages together / parts of messages.
 The application protocol must define a way to separate
- Market and the separate messages within the stream.
 Affected by congestion –
- avoidance mechanism etc.

UDP

- Unreliable
 Should consider that when working with UDP
- e.g., set a timeout when sending a query and waiting for a response
- Transfers datagrams

Word of caution - packing

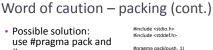
- Assume you want to have a struct represent your protocol header (or part of it)
- struct ProtocolHeader {
 unsigned short datagramLength;
 unsigned short datagramType;
 unsigned char flag;
 //...
 };

Word of caution – packing (cont.)

- Compiler may add padding to guarantee alignment
- Simply sending the struct "as-is" is not portable
- Output:
- 0 4 8 16
- S's size is: 24

st	ruct S { short i:	//2 bytes
	int j;	//4 bytes
	char k:	//1 byte
	double l;	//8 bytes
};		
in {	printf("%ld " printf("%ld " printf("%ld\r	, offsetof(S, i)); , offsetof(S, j)); , offsetof(S, k)); \n", offsetof(S, l)); te is: %ld\r\n\r\n", sizeof(S));
}		

#include <stdio.h> #include <stddef.h>



- #pragma pop Code portability issues
- Output: 0267
 - T's size is: 15

#pragma pack(push, 1)
struct T {
 short i; //2 bytes
 int j;
 char k;
 double l; //4 bytes //1 byte //8 bytes

#pragma pack(pop)

int main()

printf("%id ", offsetof{T, i}); printf("%id ", offsetof{T, j}); printf("%id ", offsetof{T, k}); printf("%id/\n", offsetof(T, l); printf(""i/\n", i?; size i; %id/r\n\r\n", sizeof{T});

Socket Programming – Part II

Handling blocking calls

Blocking function calls

- · Many of the functions we saw block until a certain event
- accept: until a client initiates a session
- · connect: until the connection is (half) established
- recv. recvfrom: until a data is received
- · send, sendto: until data is pushed into the socket's buffer
- · For simple programs, blocking is convenient
- What about more complex programs?
 - multiple connections
 - simultaneous sends and receives
 - simultaneously doing non-networking processing

How do we handle blocking?

- Initiate multiple threads
- Do not allow blocking by the use of fcntl()
- · Call a function only when it's guaranteed not to block
 - select(), pselect(), poll(), ppoll()
 - select() gets a set of fd's and returns which of them is
 - Read-ready: recv() (data socket) or accept() (listening socket) will not block
 - Write-ready: send() will not block

select()

- int select(int nfds, fd set *readfds, fd set *writefds, fd set *exceptfds, struct timeval *timeout);
- nfds: highest-numbered file descriptor in any of the three sets, plus 1.
- readfds, writefds, exceptfds: sets of fd's to see if they're read-ready, write-ready or except-ready
 - " "Exceptional conditions" are not errors, but rather states of the sockets (e.g. TCP's urgent ptr is set).
 - Any set can be replaced with NULL \rightarrow the corresponding condition will not be checked.

select() (cont.)

- Returns when at least one of the watched fd's becomes ready, or when the timeout expires
 - Returns the total number of ready fd's in all the sets.
 The sets are changed to indicate which fd's are ready.
 - Returns 0 if timeout expired
 - Returns -1 on error (and errno is set accordingly).

Working with fd_set

- fd_set is just a bit vector
- void FD_ZERO (fd_set *set)
 Initializes to an empty set
- void FD_SET (int fd, fd_set *set)
 Adds fd to the set
- int **FD_ISSET** (int fd, fd_set *set) • Returns non-zero value if fd is in the set, 0 otherwise
- void FD_CLR (int fd, fd_set *set)
 Removes fd from the set
- stdin, stdout, stderr are associated with fd's 0, 1, 2 respectively

select's timeout argument

```
struct timeval {
   long tv_sec; /* seconds */
   long tv_usec; /* microseconds, always less
   than 10^6 */
```

```
};
```

- Pass (0,0) to return immediately
- Pass NULL pointer to wait indefinitely until one of the fd's is ready
- Some OS's decrease the time elapsed, some don't
 Linux does

fd_set read_fds;
<pre>// main loop of the program for(;;) { Fl_ZERO(sread_fds); //reset fd set FD_SET(listening=sock, sread_fds); for(/* for each active client with fd = client_sock */) { FD_SET(client_sock, sread_fds); } fdmax = // the highest fd in read_fds</pre>
<pre>select(fdmax + 1, &read_fds, NULL, NULL, NULL);</pre>
<pre>if (FD_ISSET(listening_sock , &read_fds)) { // listening socket is read-ready: a new client is available. // new_client_sock = accept(listening_sock,)</pre>
<pre>for(/* for each active client with fd = client_sock */) { if (FD_ISSET(client_sock , sread_fds)) { // client socket is read ready - unread data is available // nbytes = recv(client_sock, _ } } //END main program loop</pre>

select example: reading from multiple active sockets