JYOTHISHMATHI INSTITUTE OF TECHNOLOGY & SCIENCE

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DEPARTMENT OF CSE SUBJECT: Data Communication And Computer Networks

NAME OF THE FACULTYB.SUSHMITHA YEAR / SEMESTER: III-I CSE-A,B&C

TCP/IP model

- Data Link Layer protocol
- □ Ethernet (IEEE 802.3) is widely used.
- Supported by a variety of physical layer implementations.
- Multi-access (shared medium).

CSMA/CD

- Carrier Sense Multiple Access with Collision Detection
- Carrier Sense: can tell when another host is transmitting
- □ Multiple Access: many hosts on 1 wire
- □ Collision Detection: can tell when another host transmits at the same time.

An Ethernet Frame



- □ The preamble is a sequence of alternating
 1s and 0s used for synchronization.
- □ CRC is Cyclic Redundency Check

Ethernet Addressing

- □ Every Ethernet interface has a unique 48 bit address (a.k.a. hardware address).
 - Example: C0:B3:44:17:21:17
 - The broadcast address is all 1's.
 - Addresses are assigned to vendors by a central authority.
- Each interface looks at every frame and inspects the destination address. If the address does not match the hardware address of the interface (or the broadcast address), the frame is discarded.

Internet Protocol

- □ IP is the network layer
 - packet delivery service (host-to-host).
 - translation between different data-link protocols
- □ IP provides connectionless, unreliable delivery of IP datagrams.
 - Connectionless: each datagram is independent of all others.
 - Unreliable: there is no guarantee that datagrams are delivered correctly or even delivered at all.

IP Addresses

- IP addresses are not the same as the underlying data-link (MAC) addresses.
- □ IP is a network layer it must be capable of providing communication between hosts on different kinds of networks (different datalink implementations).
- □ The address must include information about what network the receiving host is on. This is what makes routing feasible.

IP Addresses

- □ IP addresses are *logical* addresses (not physical)
- □ 32 bits. IPv4 (version 4)
- □ Includes a network ID and a host ID.
- □ Every host must have a unique IP address.
- □ IP addresses are assigned by a central authority (American Registry for Internet Numbers for North America).

The four formats of IP Addresses

Class

A 0 NetID HostID

128 possible network IDs, over 4 million host IDs per network ID

10 NetID HostID

16K possible network IDs, 64K host IDs per network ID

C NetID HostID

Over 2 million possible network IDs, 256 host IDs per network ID

Multicast Address

8 bits
8 bits
8 bits
700/TD

Network and Host IDs

- A Network ID is assigned to an organization by a global authority.
- Host IDs are assigned locally by a system administrator.
- Both the Network ID and the Host ID are used for routing.

IP Addresses

□ IP Addresses are usually shown in dotted decimal notation:

1.2.3.4

0000001 00000010 00000011 00000100

□ cse.unr.edu is 134.197.40.3

10000110 11000101 00101000 00000010



CSE has a class B network

Host and Network Addresses

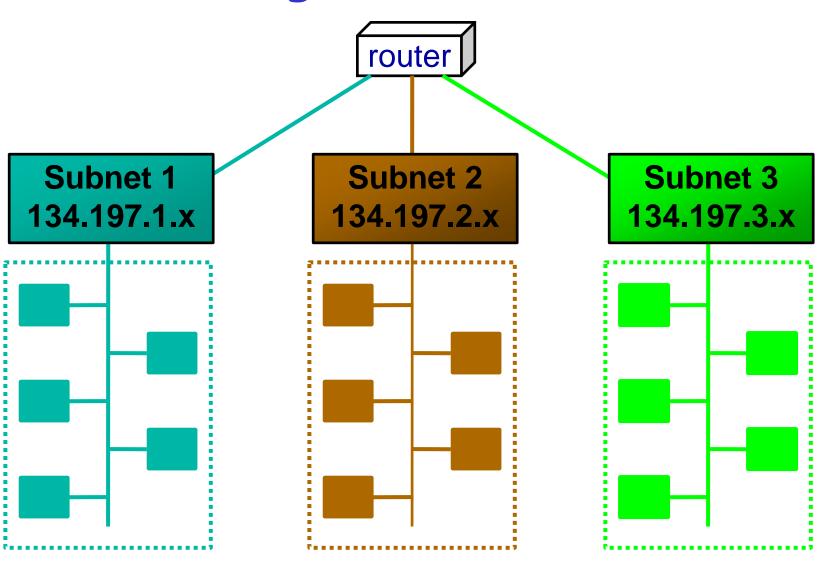
- □ A single network interface is assigned a single IP address called the host address.
- A host may have multiple interfaces, and therefore multiple host addresses.
- Hosts that share a network all have the same IP network address (the network ID).
- An IP address that has a host ID of all 0s is called a network address and refers to an entire network.

Subnet Addresses

- An organization can subdivide it's host address space into groups called subnets.
- The subnet ID is generally used to group hosts based on the physical network topology.

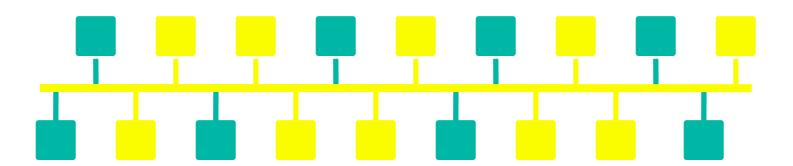


Subnetting



Subnetting

- Subnets can simplify routing.
- □ IP subnet broadcasts have a hostID of all 1s.
- It is possible to have a single wire network with multiple subnets.



Mapping IP Addresses to Hardware Addresses

- □ IP Addresses are not recognized by hardware.
- □ If we know the IP address of a host, how do we find out the hardware address?
- □ The process of finding the hardware address of a host given the IP address is called

Address Resolution

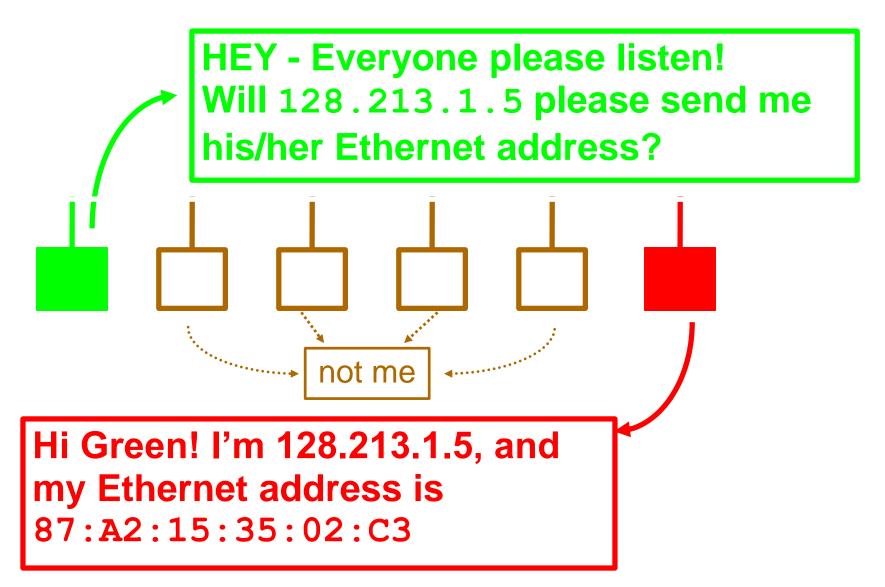
<u>ARP</u>

□ The Address Resolution Protocol used by a sending host when it knows the IP address of the destination but needs the Ethernet (or whatever) address.

Arp Arp

- ARP is a broadcast protocol every host on the network receives the request.
- Each host checks the request against it's IP address - the right one responds.
- □ hosts remember the hardware addresses of each other.

ARP conversation



IP Datagram

1 byte 1 byte 1 byte 1 byte **VERS** HL Service **Fragment Length FLAG Fragment Offset Datagram ID** TTL **Header Checksum Protocol Source Address Destination Address Options (if any) Data**

IP Datagram Fragmentation

- Packets are fragmented due to link's Maximum Transmission Unit (MTU)
- Each fragment (packet) has the same structure as the IP datagram.
- □ IP specifies that datagram reassembly is done only at the destination (not on a hop-by-hop basis).
- ☐ If any of the fragments are lost the entire datagram is discarded (and an ICMP message is sent to the sender).

IP Flow Control & Error Detection

- □ If packets arrive too fast the receiver discards excessive packets and sends an ICMP message to the sender (SOURCE QUENCH).
- ☐ If an error is found (header checksum problem) the packet is discarded and an ICMP message is sent to the sender.

ICMP Internet Control Message Protocol

- ICMP is a protocol used for exchanging control messages.
- □ ICMP uses IP to deliver messages.
- □ ICMP messages are usually generated and processed by the IP software, not the user process.

ICMP Message Types

- □ Echo Request
- Echo Response
- Destination Unreachable
- Redirect
- □ Time Exceeded
- Redirect (route change)
- □ there are more ...

Transport Layer & TCP/IP

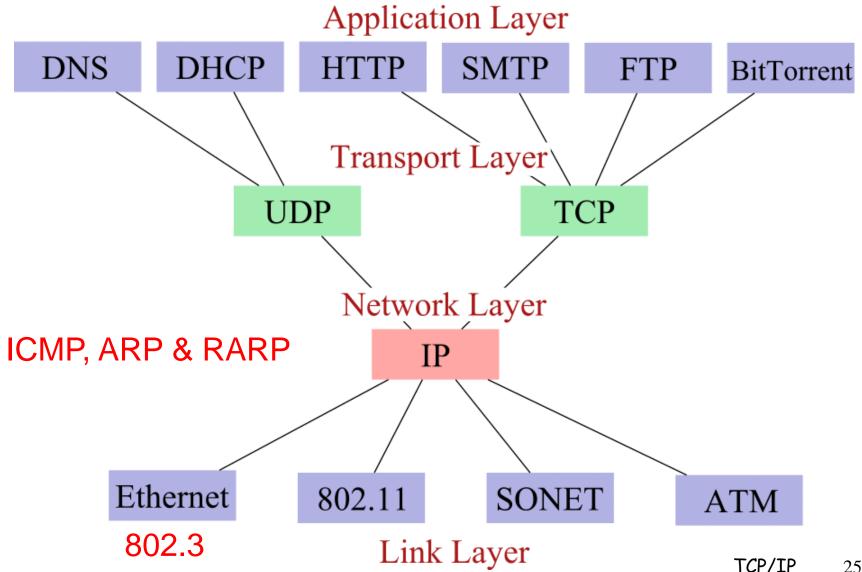
Q: We know that IP is the network layer

- so TCP must be the transport layer, right?

A: No... well, almost.

TCP is only part of the TCP/IP transport layer - the other part is UDP (User Datagram Protocol).

The Internet Hourglass



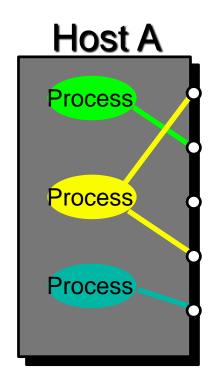
UDP User Datagram Protocol

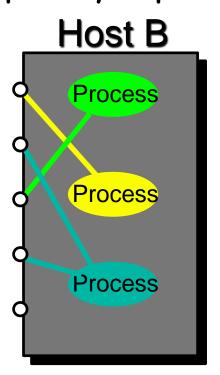
- UDP is a transport protocol
 - communication between <u>processes</u>
- UDP uses IP to deliver datagrams to the right host.

■ UDP uses ports to provide communication services to individual processes.

<u>Ports</u>

- □ TCP/IP uses an abstract destination point called a protocol port.
- □ Ports are identified by a positive integer.
- Operating systems provide some mechanism that processes use to specify a port.





<u>UDP</u>

- Datagram Delivery
- Connectionless
- Unreliable
- Minimal

UDP Datagram Format

Source Port	Destination Port
Length	Checksum
Data	

TCP Transmission Control Protocol

- □ TCP is an alternative transport layer protocol supported by TCP/IP.
- □ TCP provides:
 - Connection-oriented
 - *Reliable
 - *Full-duplex
 - *Byte-Stream

Connection-Oriented

- Connection oriented means that a virtual connection is established before any user data is transferred.
- ☐ If the connection cannot be established, the user program is notified (finds out).
- ☐ If the connection is ever interrupted, the user program(s) is finds out there is a problem.

Reliable

- □ Reliable means that every transmission of data is acknowledged by the receiver.
- Reliable does not mean that things don't go wrong, it means that we find out when things go wrong.
- □ If the sender does not receive acknowledgement within a specified amount of time, the sender retransmits the data.

Byte Stream

Stream means that the connection is treated as a stream of bytes.

□ The user application does not need to package data in individual datagrams (as with UDP).

Buffering

- TCP is responsible for buffering data and determining when it is time to send a datagram.
- It is possible for an application to tell TCP to send the data it has buffered without waiting for a buffer to fill up.

Full Duplex

- □ TCP provides transfer in both directions (over a single virtual connection).
- □ To the application program these appear as 2 unrelated data streams, although TCP can piggyback control and data communication by providing control information (such as an ACK) along with user data.

TCP Ports

- □ Interprocess communication via TCP is achieved with the use of ports (just like UDP).
- □ UDP ports have no relation to TCP ports (different name spaces).

TCP Segments

- □ The chunk of data that TCP asks IP to deliver is called a TCP segment.
- □ Each segment contains:
 - data bytes from the byte stream
 - control information that identifies the data bytes

TCP Segment Format

1 byte 1 byte 1 byte 1 byte **Source Port Destination Port Sequence Number Request Number** offset Reser. **Control** Window Checksum **Urgent Pointer Options (if any)** Data

Addressing in TCP/IP

- □ Each TCP/IP address includes:
 - Internet Address
 - Protocol (UDP or TCP)
 - Port Number

NOTE: TCP/IP is a protocol suite that includes IP, TCP and UDP

TCP vs. UDP

Q: Which protocol is better?

A: It depends on the application.

TCP provides a connection-oriented, reliable, byte stream service (lots of overhead).

UDP offers minimal datagram delivery service (as little overhead as possible).

TCP Lingo

When a client requests a connection, it sends a "SYN" segment (a special TCP segment) to the server port.

SYN stands for synchronize. The SYN message includes the client's ISN.

□ ISN is Initial Sequence Number.

More...

- □ Every TCP segment includes a Sequence Number that refers to the first byte of data included in the segment.
- Every TCP segment includes a Request Number (Acknowledgement Number) that indicates the byte number of the next data that is expected to be received.
 - All bytes up through this number have already been received.

And more...

- □ There are a bunch of control flags:
 - URG: urgent data included.
 - ACK: this segment is (among other things) an acknowledgement.
 - * RST: error abort the session.
 - SYN: synchronize Sequence Numbers (setup)
 - * FIN: polite connection termination.

And more...

□ MSS: Maximum segment size (A TCP option)

■ Window: Every ACK includes a Window field that tells the sender how many bytes it can send before the receiver will have to toss it away (due to fixed buffer size).

TCP Connection Creation

□ Programming details later - for now we are concerned with the actual communication.

- □ A server accepts a connection.
 - Must be looking for new connections!
- □ A client requests a connection.
 - * Must know where the server is!

Client Starts

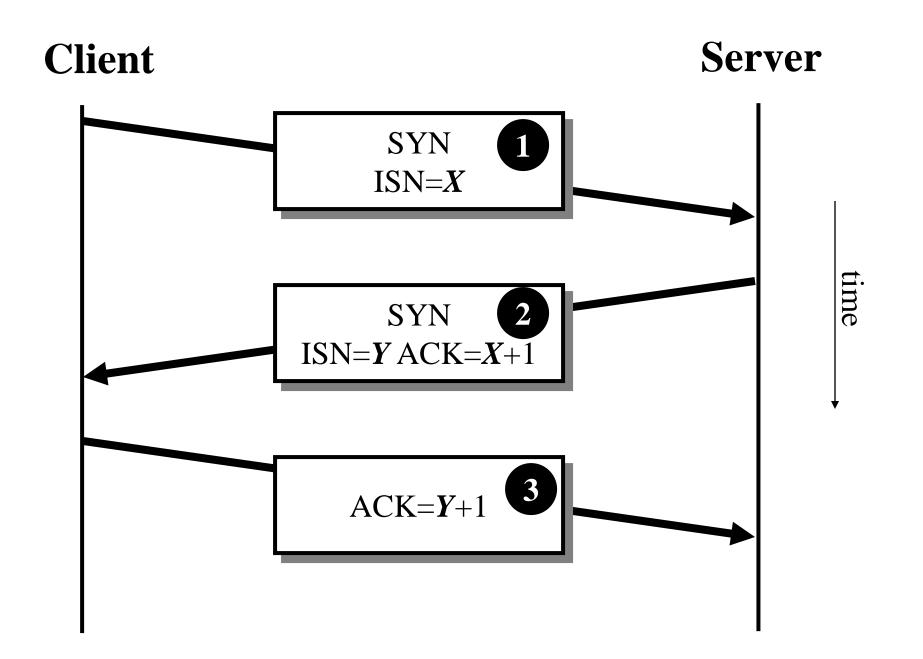
- A client starts by sending a SYN segment with the following information:
 - Client's ISN (generated pseudo-randomly)
 - Maximum Receive Window for client.
 - Optionally (but usually) MSS (largest datagram accepted).
 - No payload! (Only TCP headers)

Sever Response

- When a waiting server sees a new connection request, the server sends back a SYN segment with:
 - Server's ISN (generated pseudo-randomly)
 - Request Number is Client ISN+1
 - Maximum Receive Window for server.
 - Optionally (but usually) MSS
 - No payload! (Only TCP headers)

Finally

- When the Server's SYN is received, the client sends back an ACK with:
 - Request Number is Server's ISN+1



TCP Data and ACK

- Once the connection is established, data can be sent.
- Each data segment includes a sequence number identifying the first byte in the segment.
- Each segment (data or empty) includes a request number indicating what data has been received.

TCP Buffers

- □ The TCP layer doesn't know when the application will ask for any received data.
 - TCP buffers incoming data so it's ready when we ask for it.
- Both the client and server allocate buffers to hold incoming and outgoing data
 - * The TCP layer does this.
- Both the client and server announce with every ACK how much buffer space remains (the Window field in a TCP segment).

Send Buffers

□ The application gives the TCP layer some data to send.

- □ The data is put in a send buffer, where it stays until the data is ACK'd.
 - it has to stay, as it might need to be sent again!
- □ The TCP layer won't accept data from the application unless (or until) there is buffer space.

<u>ACKs</u>

- □ A receiver doesn't have to ACK every segment (it can ACK many segments with a single ACK segment).
- Each ACK can also contain outgoing data (piggybacking).
- □ If a sender doesn't get an ACK after some time limit (MSL) it resends the data.

TCP Segment Order

- Most TCP implementations will accept out-oforder segments (if there is room in the buffer).
- □ Once the missing segments arrive, a single ACK can be sent for the whole thing.
- □ Remember: IP delivers TCP segments, and IP in not reliable IP datagrams can be lost or arrive out of order.

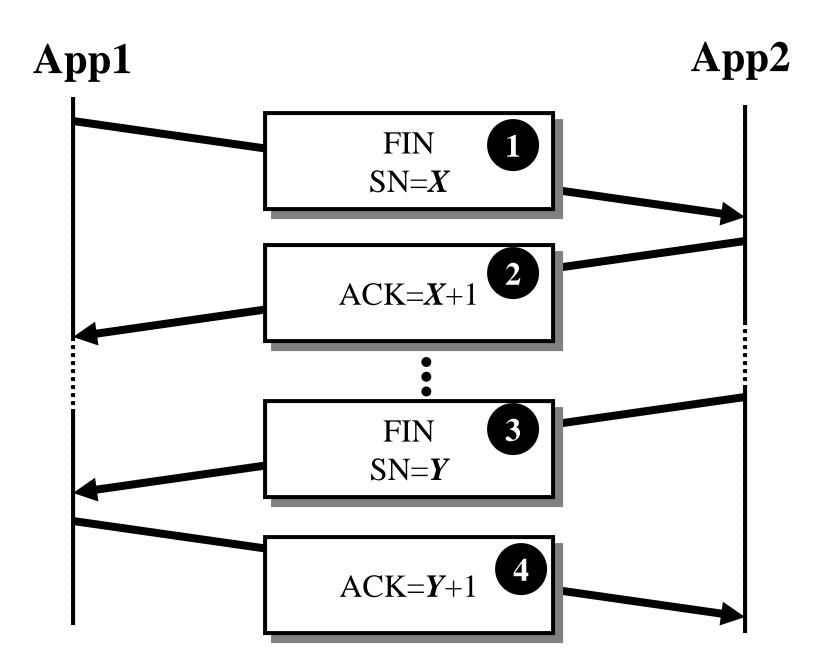
Termination

□ The TCP layer can send a RST segment that terminates a connection if something is wrong.

■ Usually the application tells TCP to terminate the connection politely with a FIN segment.

<u>FIN</u>

- □ Either end of the connection can initiate termination.
- □ A FIN is sent, which means the application is done sending data.
- ☐ The FIN is ACK'd.
- □ The other end must now send a FIN.
- ☐ That FIN must be ACK'd.



TCP Termination

- 1 App1: "I have no more data for you".
- 2 App2: "OK, I understand you are done sending." dramatic pause...
- 3 App2: "OK Now I'm also done sending data".
- App1: "Roger, Over and Out, Goodbye, Astalavista Baby, Adios, It's been real ..."

camera fades to black ...

TCP TIME_WAIT

- Once a TCP connection has been terminated (the last ACK sent) there is some unfinished business:
 - What if the ACK is lost? The last FIN will be resent and it must be ACK'd.
 - What if there are lost or duplicated segments that finally reach the destination after a long delay?
- □ TCP hangs out for a while to handle these situations.

Test Questions

- Why is a 3-way handshake necessary?
 - * HINTS: TCP is a reliable service, IP delivers each TCP segment, IP is not reliable.
- Who sends the first FIN the server or the client?
- Once the connection is established, what is the difference between the operation of the server's TCP layer and the client's TCP layer?
- □ What happens if a bad guy can guess ISNs?