CSE 461 – Module 11

Connections

This Time

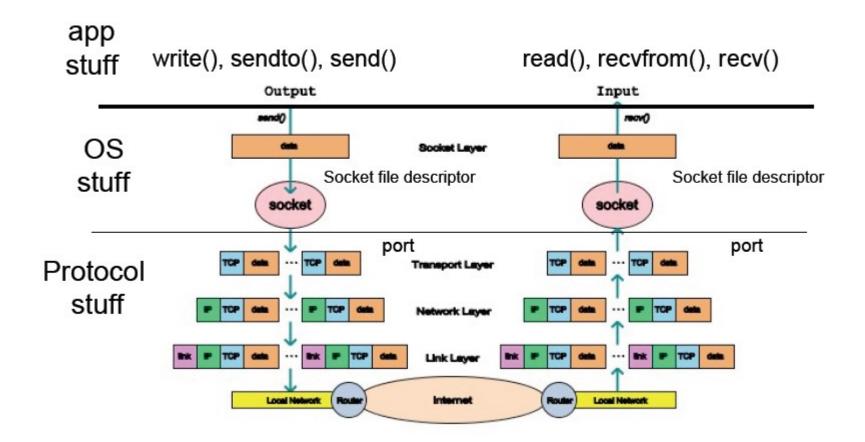
- More on the Transport Layer
- Focus
 - How do we <u>connect processes</u>?
- Topics
 - Naming processes
 - Connection setup / teardown
 - Flow control

Application Presentation Session Transport Network Data Link Physical

Naming Processes/Services

- Process here is an abstract term for your Web browser (HTTP), Email servers (SMTP), hostname translation (DNS), mp3 player (RTSP), etc.
- How do we identify for remote communication?
 - Process id or memory address are OS-specific and transient
- So TCP and UDP use Ports
 - 16-bit integers representing mailboxes that processes "rent"
 - typically from OS
 - Identify endpoint uniquely as (IP address, protocol, port)
 - OS converts into process-specific channel, like "socket"

Processes as Endpoints

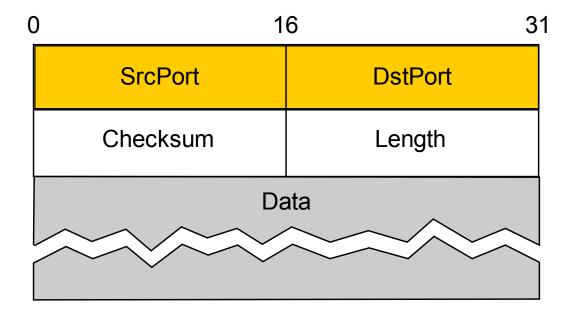


Picking Port Numbers

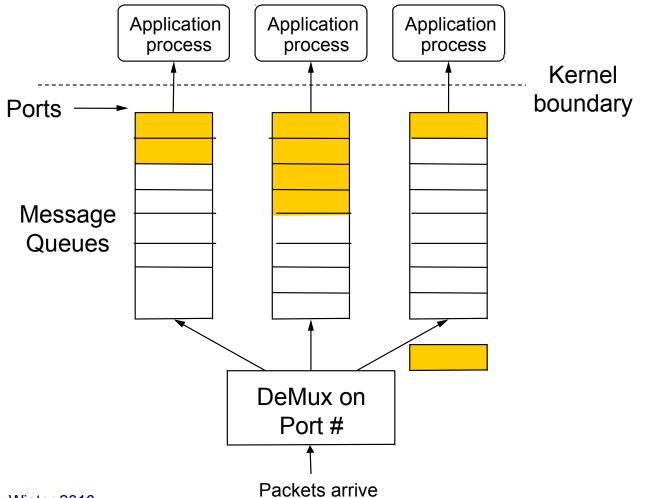
- We still have the problem of allocating port numbers
 - What port should a Web server use on host X?
 - To what port should you send to contact that Web server?
- Servers typically bind to "well-known" port numbers
 - e.g., HTTP 80, SMTP 25, DNS 53, ... look in /etc/services
 - Ports below 1024 reserved for "well-known" services
- Clients use OS-assigned temporary (ephemeral) ports
 - Above 1024, recycled by OS when client finished

User Datagram Protocol (UDP)

- Provides message delivery between processes
 - Source port filled in by OS as message is sent
 - Destination port identifies UDP delivery queue at endpoint



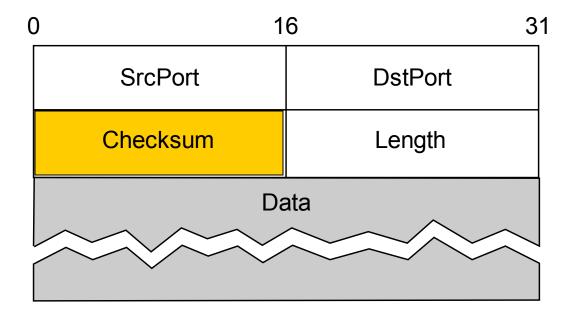
UDP Delivery



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UDP Checksum

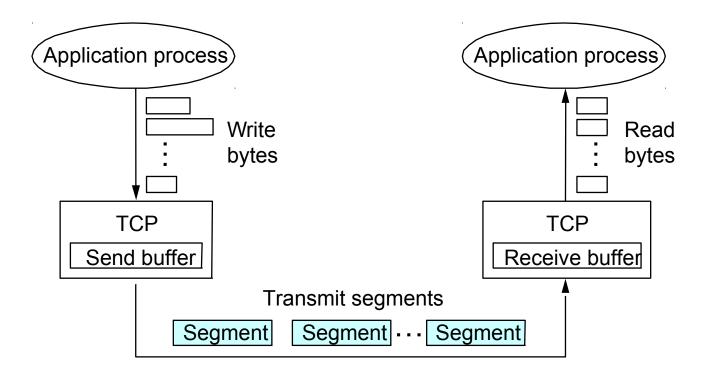
- UDP includes optional protection against errors
 - Checksum intended as an end-to-end check on delivery
 - So it covers data, UDP header, and IP pseudoheader



Transmission Control Protocol (TCP)

- Reliable bi-directional bytestream between processes
 - Message boundaries are not preserved
- Connections
 - Conversation between endpoints with beginning and end
- Flow control
 - Prevents sender from over-running receiver buffers
- Congestion control
 - Prevents sender from over-running network buffers

TCP Delivery



• Ports plus IP addresses identify a connection/flow

0	4 1	10 1	6 <u>3</u>		
SrcPort			DstPort		
SequenceNum					
Acknowledgment					
HdrLer	HdrLen 0 Flags AdvertisedWindow				
Checksum			UrgPtr		
Options (variable)					

• Sequence, Ack numbers used for the sliding window

0	4 1	0 1	6 3 [°]			
SrcPort			DstPort			
	SequenceNum					
Acknowledgment						
HdrLer	n 0	Flags	AdvertisedWindow			
Checksum			UrgPtr			
Options (variable)						

• Flags may be URG, ACK, PUSH, RST, SYN, FIN

0	4 1	0 1	6	31	
SrcPort			DstPort		
SequenceNum					
Acknowledgment					
HdrLen 0 Flags AdvertisedWindow					
Checksum			UrgPtr		
Options (variable)					
Data					

• Advertised window is used for flow control

0	2	4 1	0	6		31
	SrcPort			DstPort		
	SequenceNum					
	Acknowledgment					
	HdrLen 0 Flags			AdvertisedWindow		
	Checksum				UrgPtr	
	Options (variable)					
	Data					

TCP Connection Establishment

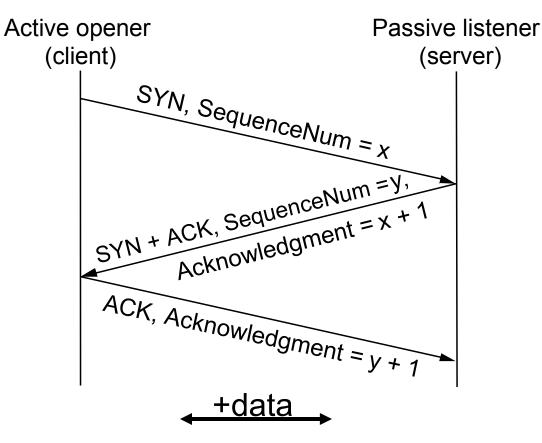
- Both connecting and closing are (slightly) more complicated than you might expect
- That they *can* work is reasonably straightforward
- Harder is what to do when things go wrong
 - TCP SYN+ACK attack
- Close looks a bit complicated because both sides have to close to be done
 - Conceptually, there are two one-way connections
 - Don't want to hang around forever if other end crashes

TCP Connection Establishment

- Both sender and receiver must be ready before we start to transfer the data
 - Sender and receiver need to agree on a set of parameters
 - e.g., the Maximum Segment Size (MSS)
- This is "signaling"
 - It sets up state at the endpoints
 - Compare to "dialing" in the telephone network
- In TCP a Three-Way Handshake is used

Three-Way Handshake

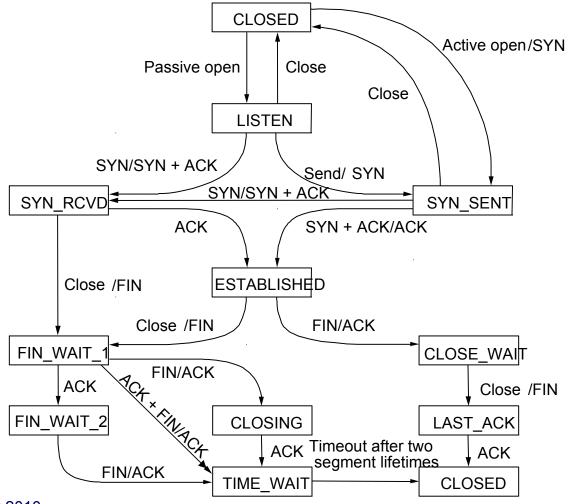
Opens both directions for transfer



Some Comments

- We could abbreviate this setup, but it was chosen to be robust, especially against delayed duplicates
 - Three-way handshake from Tomlinson 1975
- Choice of changing initial sequence numbers (ISNs) minimizes the chance of hosts that crash getting confused by a previous incarnation of a connection
- But with random ISN it actually "proves" that two hosts can communicate
 - Weak form of authentication

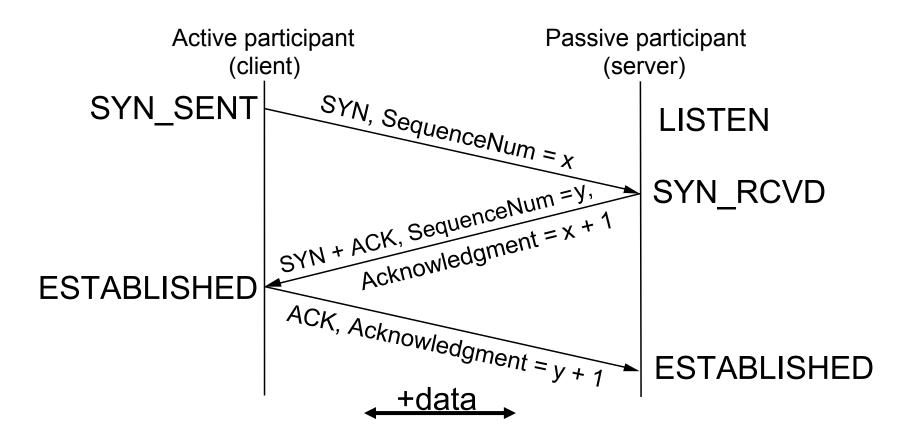
TCP State Transitions



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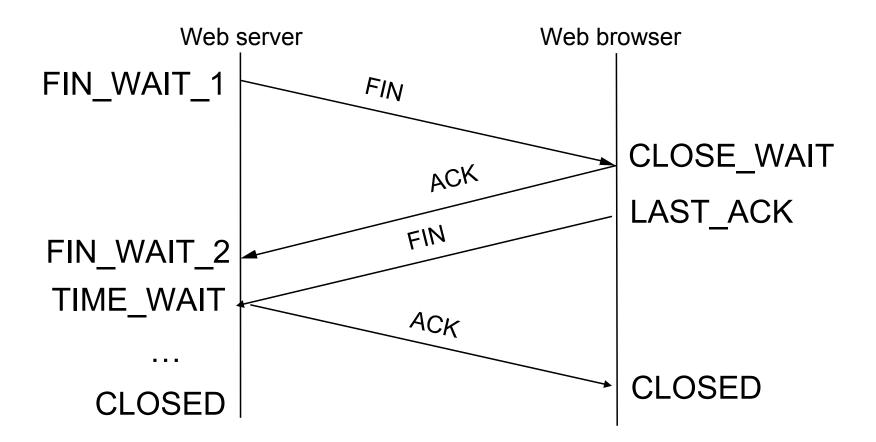
Again, with States



Connection Teardown

- Orderly release by sender and receiver when done
 - Delivers all pending data and "hangs up"
- Cleans up state in sender and receiver
- TCP provides a "symmetric" close
 - both sides shutdown independently

TCP Connection Teardown



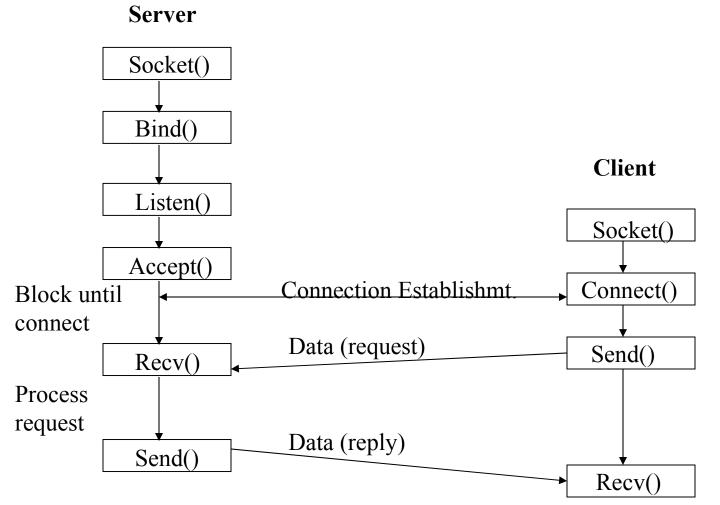
The TIME_WAIT State

- We wait 2MSL (two times the maximum segment lifetime of 60 seconds) before completing the close
- Why?
- ACK might have been lost and so FIN will be resent
- Could interfere with a subsequent connection

Berkeley Sockets interface

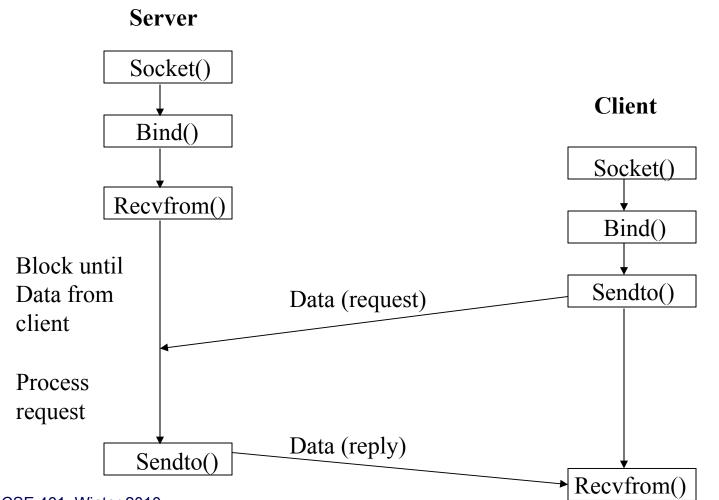
- Networking protocols implemented in OS
 - OS must expose a programming API to applications
 - most OSs use the "socket" interface
 - originally provided by BSD 4.1c in ~1982.
- Principle abstraction is a "socket"
 - a point at which an application attaches to the network
 - defines operations for creating connections, attaching to network, sending and receiving data, closing connections

TCP (connection-oriented)



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UDP (connectionless)



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Using Sockets: UDP

- import java.net.*;
- UDP sockets:
 - new DatagramSocket(); // binds to ephemeral port number
 - new DatagramSocket(port); // tries to bind to `port'
- DatagramPacket
 - Unit of transfer between application and networking software
 - new DatagramPacket(byte[] buf, int len);
 - new DatagramPacket(byte[] buf, int len, InetAddress addr, int port);
- Sending data:
 - Construct a DatagramPacket
 - Set its data field, and its address components
 - myDatagramSocket.send(myDatagramPacket)

Java / UDP

- Java also has an interface supporting *connect* (SocketAddr addr), but it's a layer above UDP
 - Filters incoming packets not from *addr*
 - Filters outgoing packets not to *addr*
- Performance / correctness issue:
 - Is a copy of the data portion of a DatagramPacket made when send()
 is invoked, or is a reference to the byte[] buf kept?
- Blocking vs. non-blocking IO
 - Non-blocking options
 - 1. import java.net.*;
 - DatagramSocket.setSOTimeout(int timeout);
 - 2. import java.nio.*;
 - More general (complicated) support

Using Sockets: TCP

- The TCP distinction between passive and active open is embedded in the (typical) socket interfaces
 - There are two kinds of sockets:
 - Socket
 - ServerSocket
- Server starts, creates a ServerSocket, binds it to a local port, and listens for a client to connect
- Client starts, creates a Socket on an ephemeral port, and connects to the server socket
- As a result of the connection, the server socket creates a *new* Socket to return to the application
 - Provides a handy way to identify/name a single flow in the application code

TCP Server-side: Java

• Create:

- ServerSocket ss = new ServerSocket();
- ServerSocket ss = new ServerSocket(port);

• Listen:

- Socket s = ss.accept();

TCP Client side: Java

- Create:
 - Socket s = new Socket();

• Connect:

- s.connect(serverAddress);
- S.connect(serverAddress, timeout);

• Use:

- It's Java, the sockets support streams, the mind boggles
- BufferedReader in = new BufferedReader(new InputStreamReader(s.getInputStream()));
 - in.readLine();
- PrintWriter out = new PrintWriter(s.getOutputStream(), true);
 - Out.print(data);
- OutStream outStream = s.getOutputStream();
 - outStream.write(buf, 0, n); // byte[] buf for n bytes starting at offset 0

Blocking Operations

- read() is a *blocking* operation
- What if other side crashes?
 - No data sent
 - No FIN sent
- Solutions on these slides are for Java, but general ideas apply universally

Method 1: Timeouts

 Most language/socket interfaces will provide a way to way to say:

read N bytes, but wait no longer than T milliseconds

- On return, you either have up to N bytes or some indication that you timed out
 - Note: read(N) can often mean "read up to N" not "wait for N"
 - Note: readline(), if it's available, means "wait until you can read a \n"
- In Java, this is done by setting a *socket option*
 - socket.setSOTimeout(5); // set 5 msec. timeout
 - reader.read(buf, off, len); // wait up to 5 msec.

Method 2: Non-blocking IO

- Every language / OS will provide some way to do nonblocking IO
 - read() can be made to always return immediately, sometimes
 with an indication that it read nothing
 - A willBlock() method is probably available
 - A waitFor(datasource[]) method will be available
 - Means "block until at least one of the data sources has data available"
- In Java, these are provided by java.nio and related packages

Not A Method: Multi-threading

- Multi-threading isn't really a solution when you need nonblocking semantics
- It is a fine solution when you're willing to block, but your goal is:
 - To overlap some processing with blocking/waiting
 - To read from more than one source
- Basic problem:
 - The application (probably) can't terminate cleanly until all threads have terminated
 - The only thread that can terminate a thread is itself
 - There's no general way to wake up a thread blocked on IO

Key Concepts

- We use ports to name processes in TCP/UDP
 - "Well-known" ports are used for popular services
- Connection setup and teardown complicated by the effects of the network on messages
 - TCP uses a three-way handshake to set up a connection
 - TCP uses a symmetric disconnect