#### Lecture 26: TCP

CS 105

Spring 2021

# **OSI Network Model**

Application	HTTP, FTP, DNS ( <i>these</i> ^ are usually in libraries)		app app		
Transport	TCP, UDP		OS CPU memory		
Network	IP, ICMP (ping)		CPU [memory]		
Link	Ethernet, WiFi		controller		
Physical	wires, signal encoding		physical transmission		
(Hard to draw firm lines here)					

## **Transport Layer Protocols**

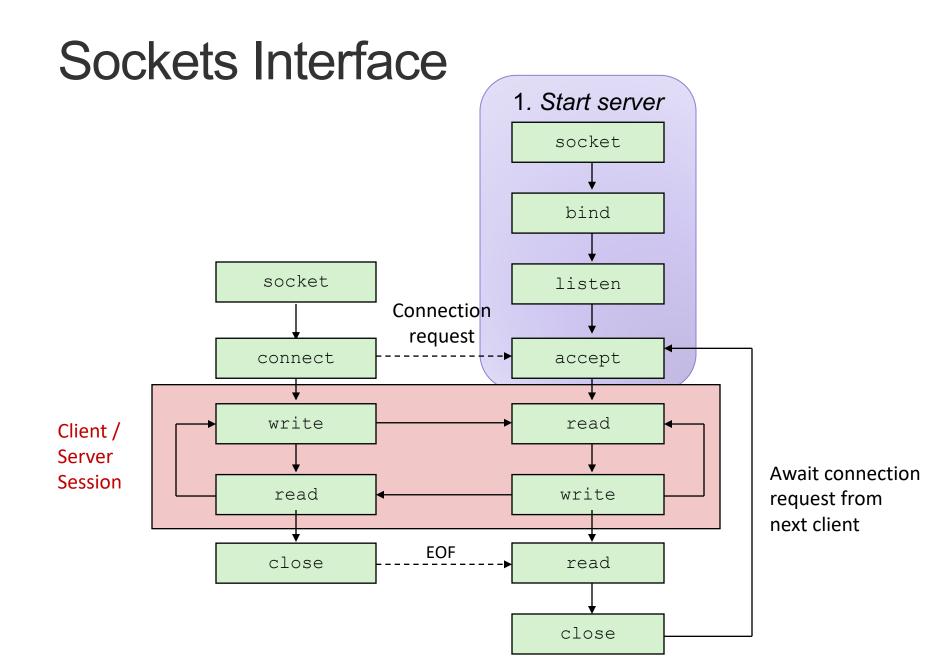
User Datagram Protocol (UDP)

- unreliable, unordered delivery
- connectionless
- best-effort, segments might be lost, delivered out-oforder, duplicated
- reliability (if required) is the responsibility of the app

Transmission Control Protocol (TCP)

reliable, in-order delivery

- connection setup
- flow control
- congestion control

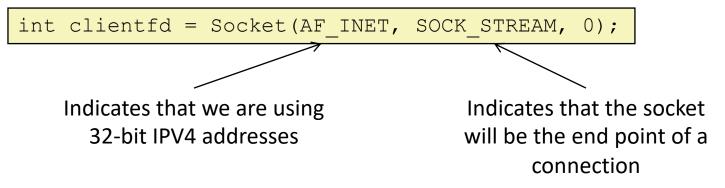


#### Sockets Interface: socket

• Clients and servers use the socket function to create a socket descriptor:

int socket(int domain, int type, int protocol)

• Example:



Protocol specific! Best practice is to use getaddrinfo to generate the parameters automatically, so that code is protocol independent.

#### Sockets Interface: bind

• A server uses bind to ask the kernel to associate the server's socket address with a socket descriptor:

int bind(int sockfd, SA \*addr, socklen\_t addrlen);

- The process can read bytes that arrive on the connection whose endpoint is addr by reading from descriptor sockfd.
- Similarly, writes to sockfd are transferred along connection whose endpoint is addr.

Best practice is to use getaddrinfo to supply the arguments addr and addrlen.

#### Sockets Interface: listen

- By default, kernel assumes that descriptor from socket function is an active socket that will be on the client end of a connection.
- A server calls the listen function to tell the kernel that a descriptor will be used by a server rather than a client:

int listen(int sockfd, int backlog);

- Converts sockfd from an active socket to a listening socket that can accept connection requests from clients.
- backlog is a hint about the number of outstanding connection requests that the kernel should queue up before starting to refuse requests.

#### Sockets Interface: accept

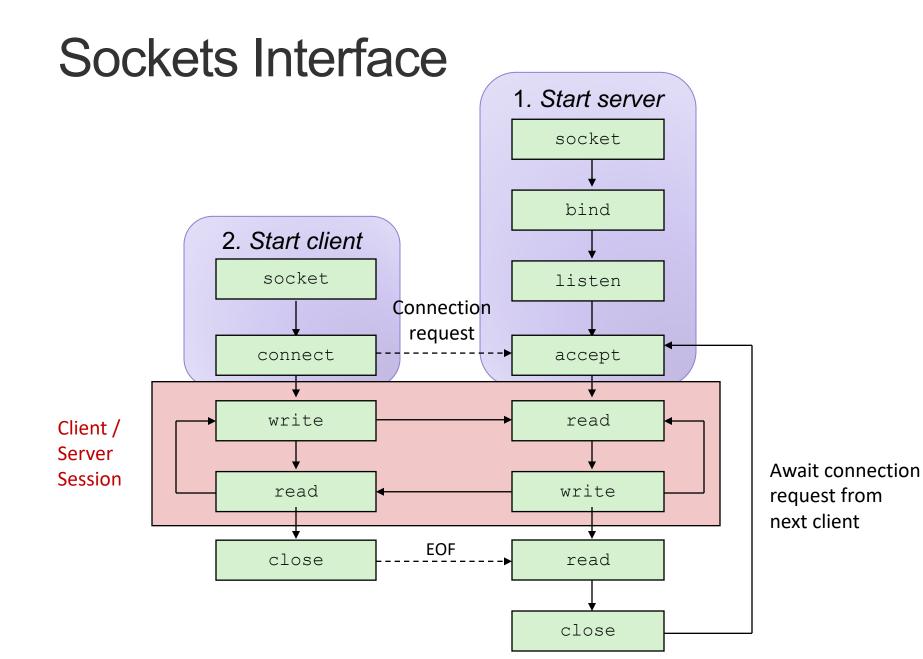
 Servers wait for connection requests from clients by calling accept:

int accept(int listenfd, SA \*addr, int \*addrlen);

- Waits for connection request to arrive on the connection bound to listenfd, then fills in client's socket address in addr and size of the socket address in addrlen.
- Returns a connected descriptor that can be used to communicate with the client via Unix I/O routines.

# Connected vs. Listening Descriptors

- Listening descriptor
  - End point for client connection requests
  - Created once and exists for lifetime of the server
- Connected descriptor
  - End point of the connection between client and server
  - A new descriptor is created each time the server accepts a connection request from a client
  - Exists only as long as it takes to service client
- Why the distinction?
  - Allows for concurrent servers that can communicate over many client connections simultaneously
    - E.g., Each time we receive a new request, we fork a child to handle the request



#### Sockets Interface: connect

 A client establishes a connection with a server by calling connect:

int connect(int clientfd, SA \*addr, socklen\_t addrlen);

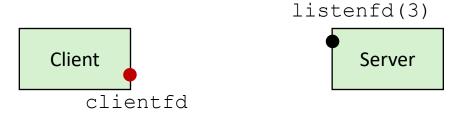
- Attempts to establish a connection with server at socket address addr
  - If successful, then clientfd is now ready for reading and writing.
  - Resulting connection is characterized by socket pair

(x:y, addr.sin\_addr:addr.sin\_port)

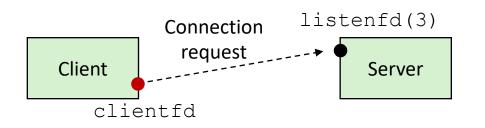
- x is client address
- y is ephemeral port that uniquely identifies client process on client host

Best practice is to use getaddrinfo to supply the arguments addr and addrlen.

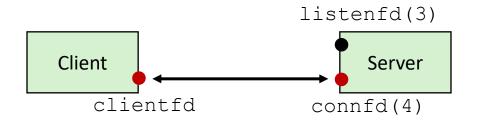
#### accept Illustrated



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



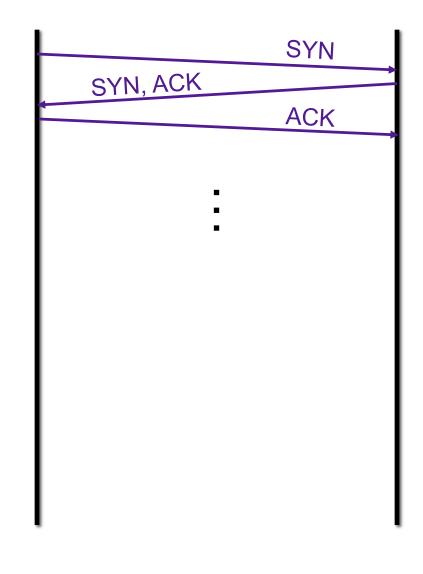
2. Client makes connection request by calling and blocking in *connect* 



3. Server returns connfd from accept. Client returns from connect. Connection is now established between clientfd and connfd

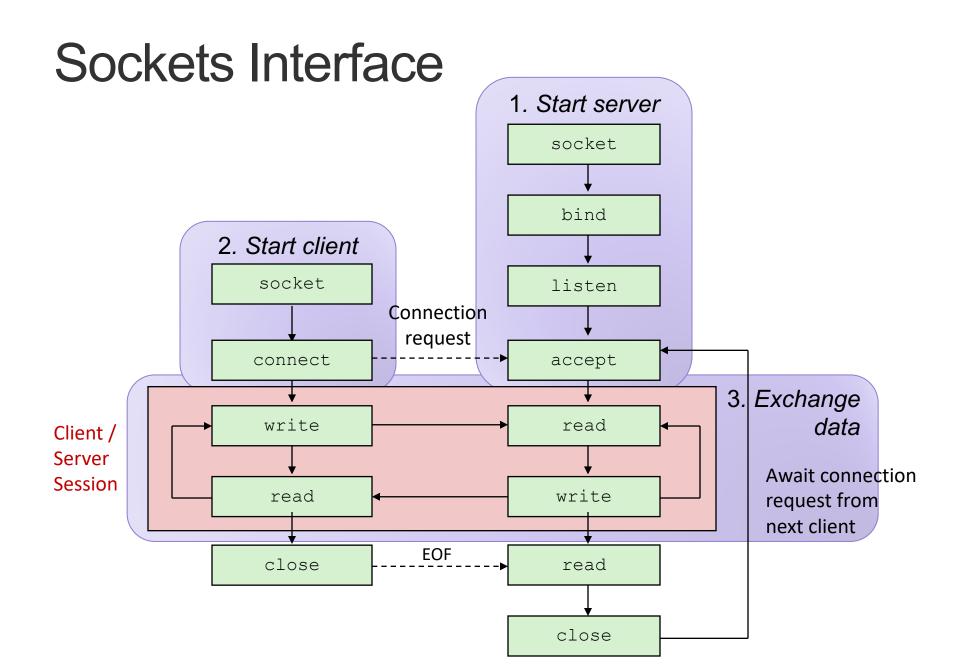
# **TCP Connections**

- TCP is connectionoriented
- A connection is initiated with a threeway handshake
- Recall: server will typically create a new socket to handle the new connection



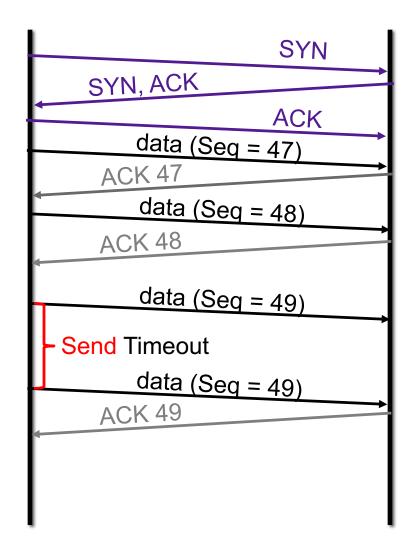
# Exercise 1: TCP Handshake

 Explain why three messages are required to set up a TCP connection



# **Reliable Transport**

- Each SYN segment will include a randomly chosen sequence number
- Sequence number of each segment is incremented by data length
- Receiver sends ACK segments acknowledging latest sequence number received
- Sender maintains copy of all sent but unacknowledged segments; resends if ACK does not arrive within timeout
- Timeout is dynamically adjusted to account for round-trip delay



# **Transport-Layer Segment Formats**

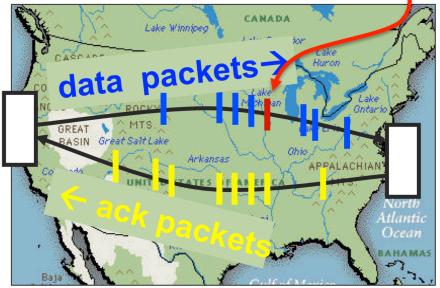
U	DP	TCP		
Source Port #	Dest. Port #	Source Port #	Dest. Port #	
		sequence number		
application me	ssage (payload)	acknowledgement number		
		HL UAPRSF	receive window	
		checksum	U data pointer	
		options		
		application message (payload)		

# **Pipelined Protocols**

- Pipelining allows sender to send multiple "in-flight", yet-tobe-acknowledged packets
  - increases throughput
  - needs buffering at sender and receiver
- how big should the window be?

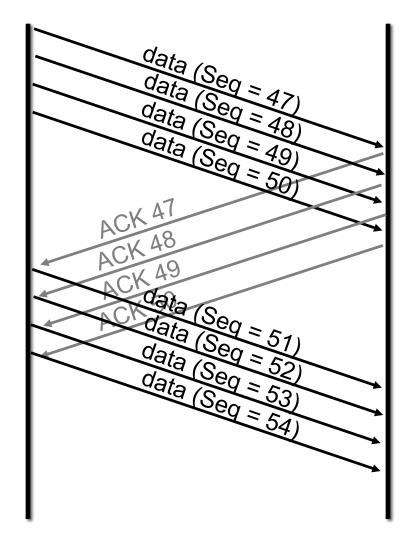
#### what if a packet in the middle goes missing?





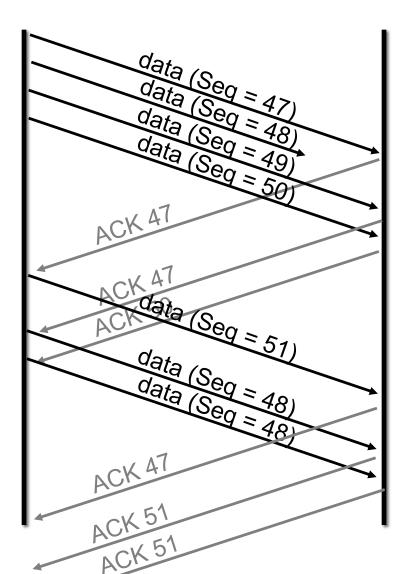
#### Example: Window Size = 4

- sender can have up to 4 unacknowledged messages
- when ACK for first message is received, it can send another message



## TCP Fast Retransmit

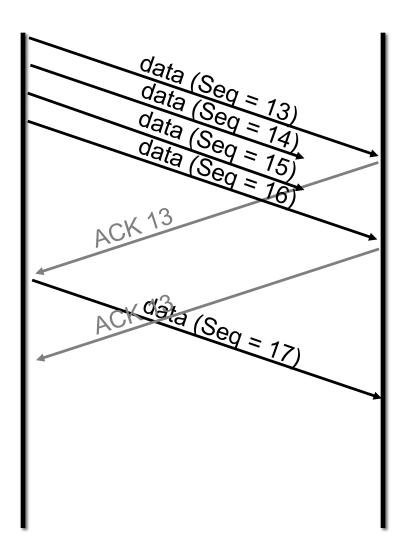
- Receiver always acks the last id it successfully received
- Sender detects loss without waiting for timeout, resends missing packet



# **Exercise 2: TCP Sequence Numbers**

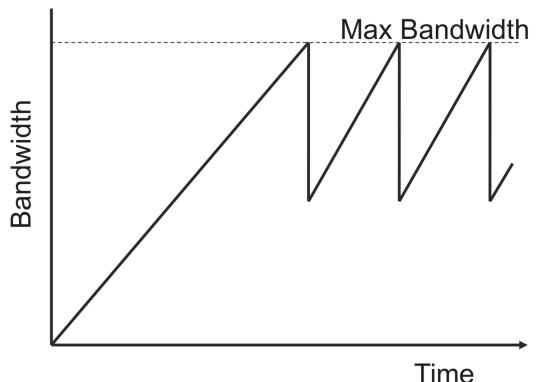
Consider the sequence of transmitted messages shown on the right

- What will be the next ACK number sent by the server?
- What will be the next Seq number sent by the client?



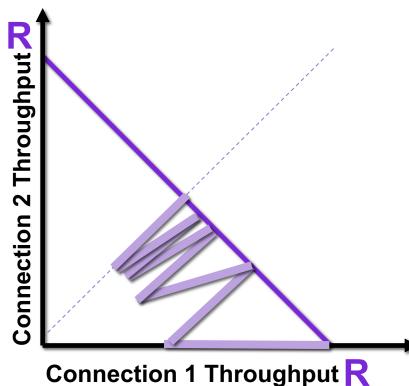
# **TCP Congestion Control**

- TCP operates under a principle of additive increasemultiplicative decrease
  - window size++ every RTT if no packets lost
  - window size/2 if a packet is dropped



# **TCP** Fairness

 Goal: if k TCP sessions share same bottleneck link of bandwidth R, each should have average rate of R/k

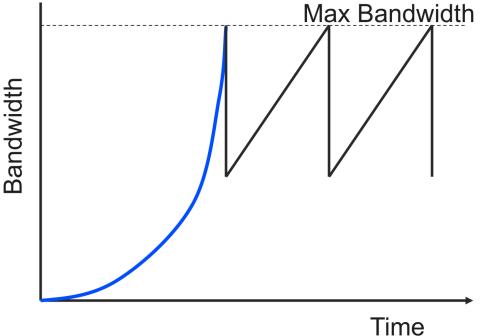


Loss: decreases throughput proportional to current bandwidth

Congestion avoidance: increases throughput linearly (evenly)

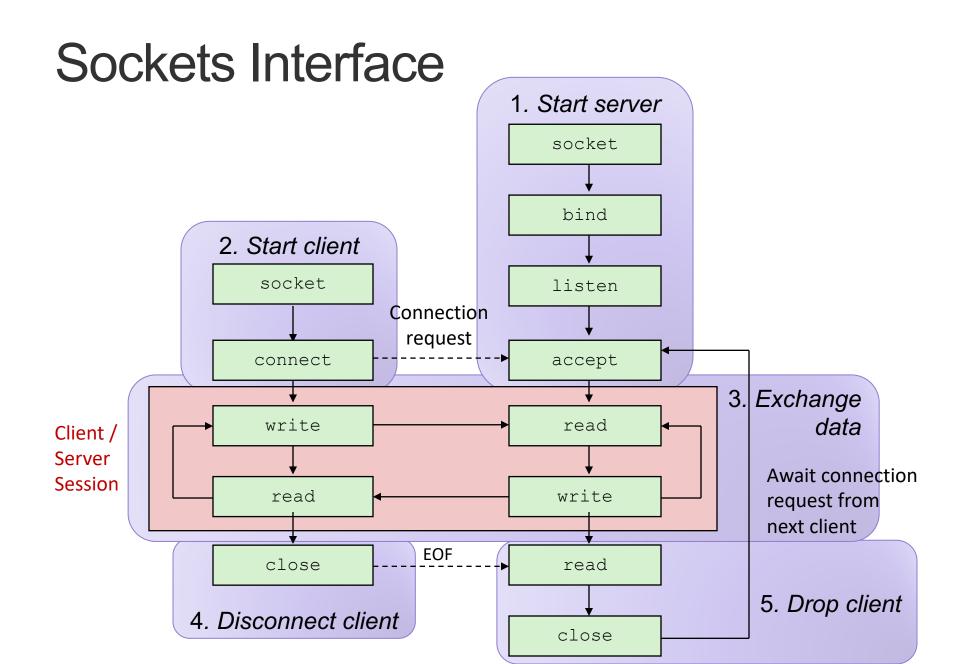
# TCP Slow Start

- Problem: linear increase takes a long time to build up a decent window size, and most transactions are small
- Solution: allow window size to increase exponentially until first loss



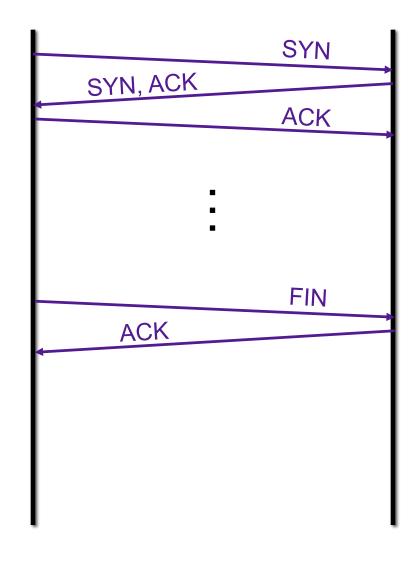
# Exercise 3: TCP Window Size

- Assume someone changes the code of their TCP client by modifying the congestion avoidance as follows: instead of increasing the window size by 1 each time an ACK is received, they double the window size each time an ACK is received (like in the slow-start phase).
- What would be the pros and cons of this modification?



# **TCP** Connections

- TCP is connectionoriented
- A connection is initiated with a three-way handshake
- Recall: server will typically create a new socket to handle the new connection
- FIN works (mostly) like SYN but to teardown a connection



# **TCP Summary**

- Reliable, in-order message delivery
- Connection-oriented, three-way handshake
- Transmission window for better throughput
  - timeouts based on link parameters (e.g., RTT, variance)
- Congestion control
  - Linear increase, exponential backoff
- Fast adaptation
  - Exponential increase in the initial phase

#### Exercise 4: Feedback

- 1. Rate how well you think this recorded lecture worked
  - 1. Better than an in-person class
  - 2. About as well as an in-person class
  - 3. Less well than an in-person class, but you still learned something
  - 4. Total waste of time, you didn't learn anything
- 2. How much time did you spend on this video lecture?
- 3. Do you have any comments or suggestions for future classes?