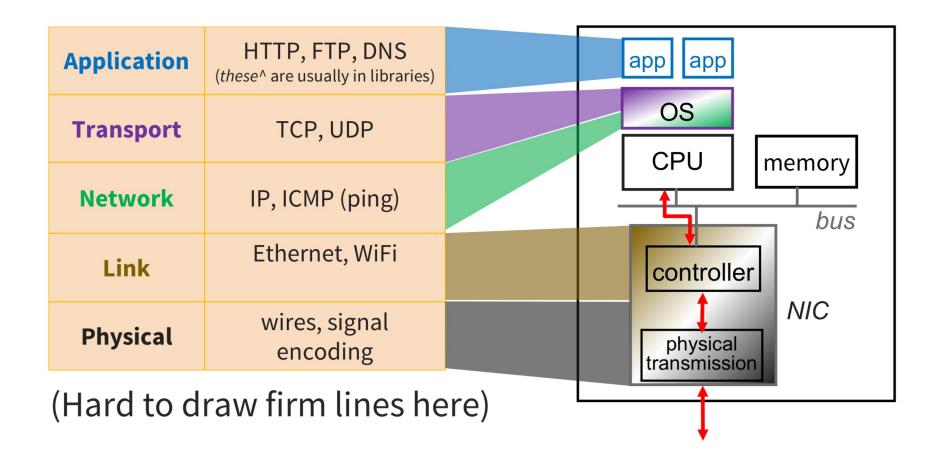
### Lecture 26: TCP

CS 105

### **OSI Network Model**



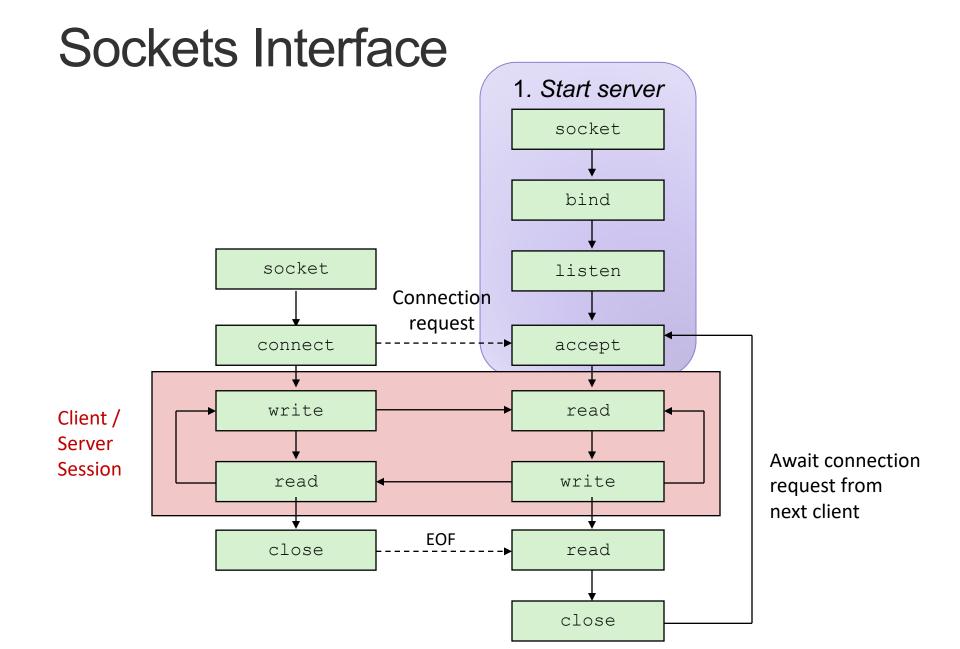
# Transport Layer Protocols

<u>User Datagram Protocol (UDP)</u>

- unreliable
- unordered delivery
- connectionless
- best-effort, segments might be lost, delivered out-oforder, duplicated
- reliability handled by app

Transmission Control Protocol (TCP)

- reliable
- in-order delivery
- connection setup
- flow control
- congestion control

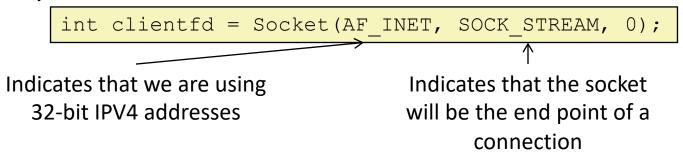


### Sockets Interface: socket

 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

#### <u>Listening</u> descriptor

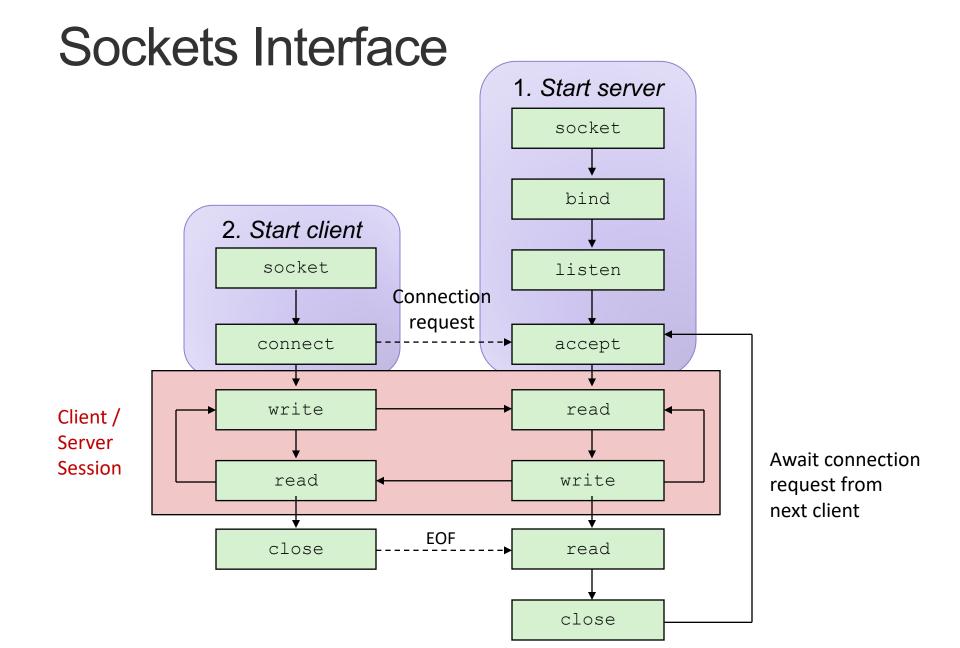
- End point for client connection requests
- Created once and exists for lifetime of the server

#### 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)
```

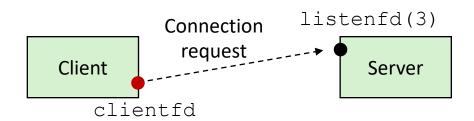
- x is client address
- y is ephemeral port that uniquely identifies client process on client host

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

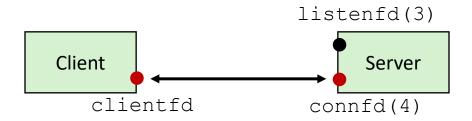
## accept Illustrated



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



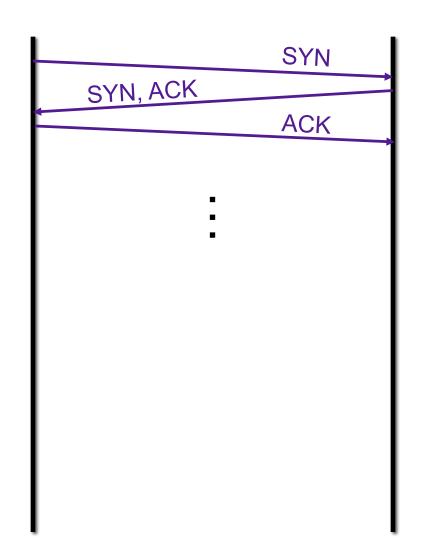
2. Client makes connection request by calling and blocking in connect



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

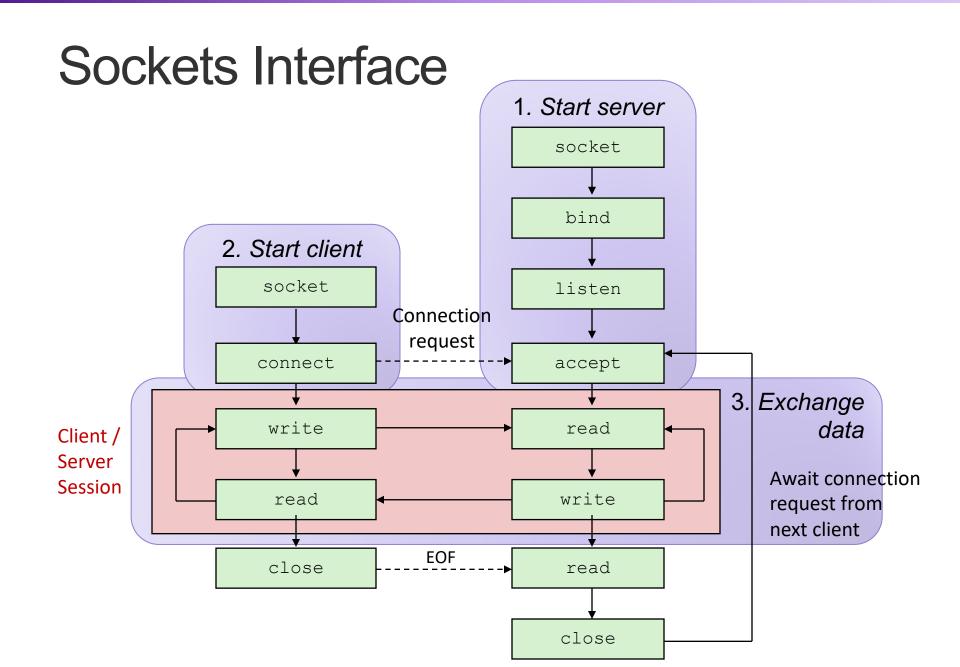
#### TCP Connections

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



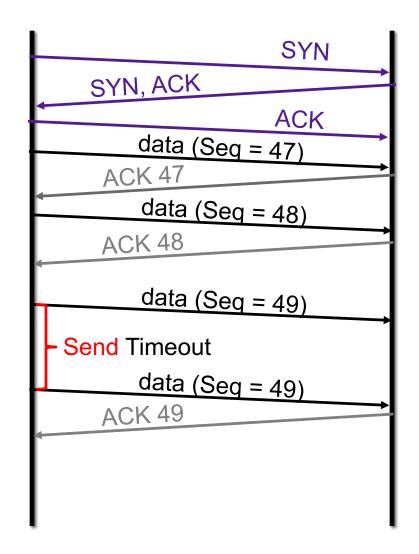
### Exercise 1: TCP Handshake

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

**UDP** 

Source Port # Dest. Port #

application message (payload)

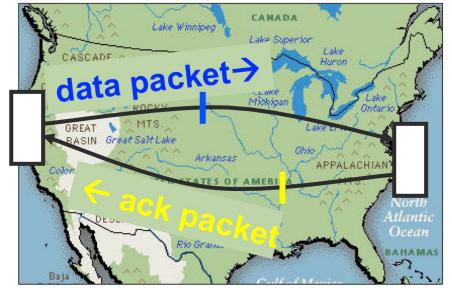
**TCP** 

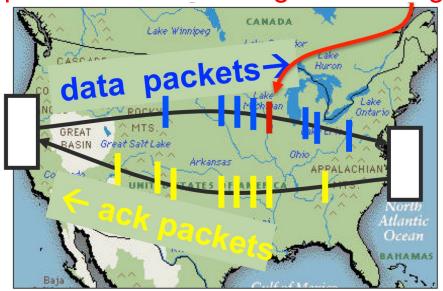
Source Port#	Dest. Port#
sequence number	
acknowledgement number	
HL UAPRSF	receive window
checksum	U data pointer
options	
application message (payload)	

# 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?

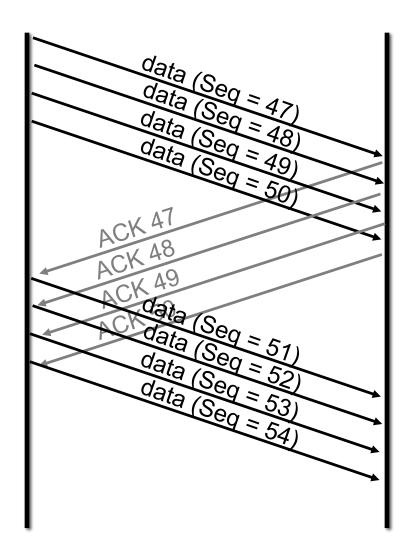




# Example: Window Size = 4

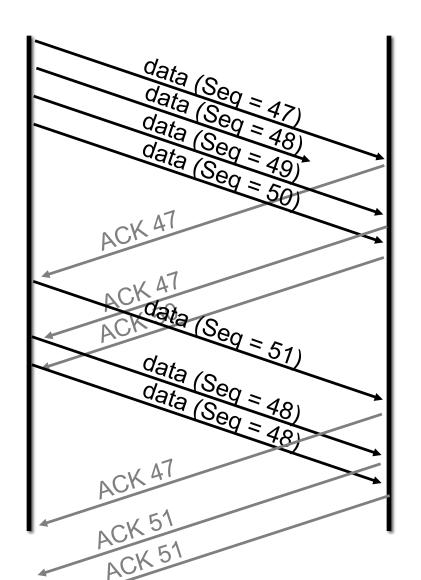
 sender can have up to 4 unacknowledged messages

 when ACK for first message is received, it can send another message



### TCP Fast Retransmit

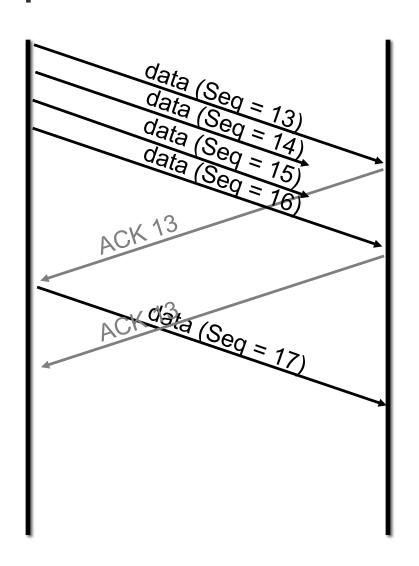
- Receiver always acks the last id it successfully received
- Sender detects loss without waiting for timeout, resends missing packet



## Exercise 2: TCP Sequence Numbers

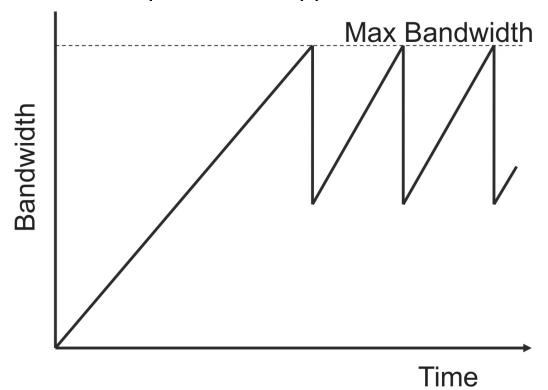
Consider the sequence of transmitted messages shown on the right

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



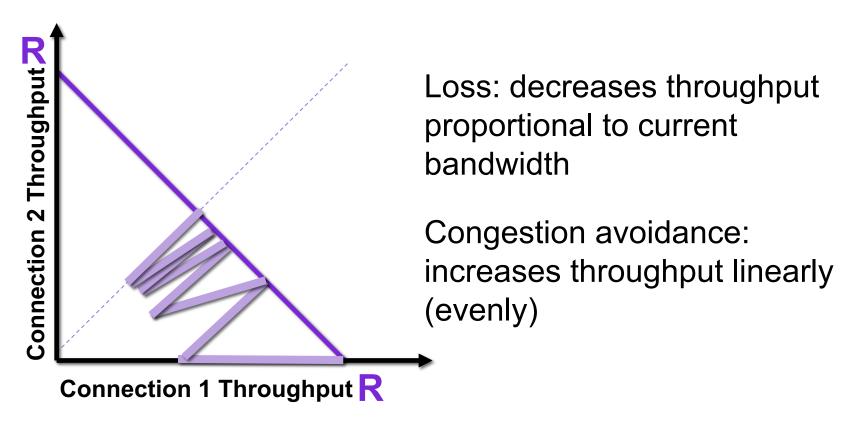
# **TCP Congestion Control**

- 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



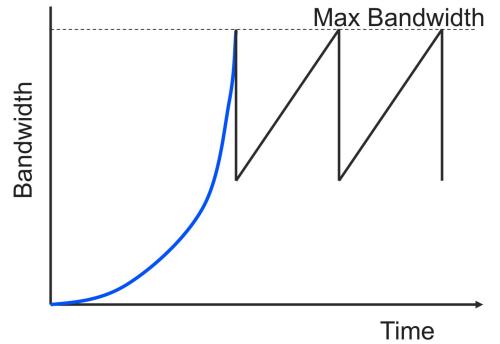
### TCP Fairness

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



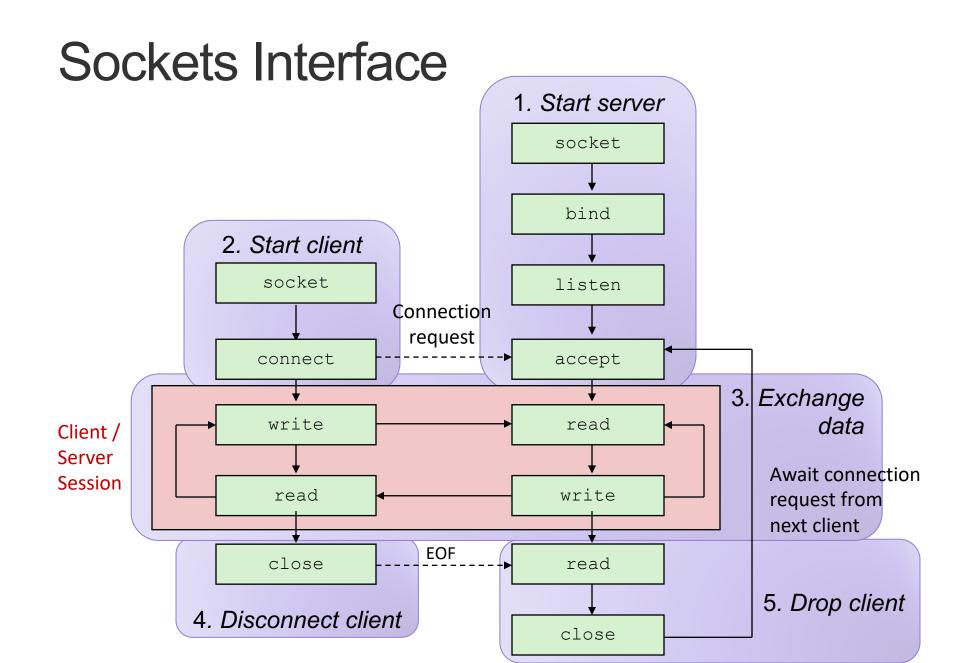
### **TCP Slow Start**

- 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



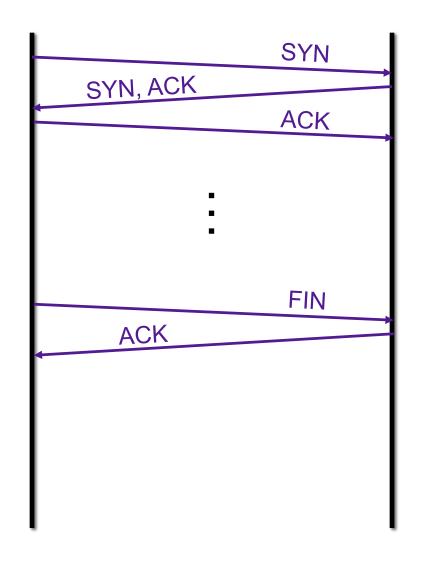
### **Exercise 3: TCP Window Size**

- 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