

# Lesson 05-03: TCP

CS 326 E Elements of Networking

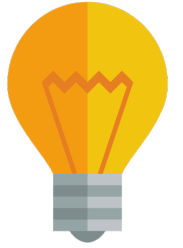
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## Example Protocols

## Responsible for

## Internet Reference Model



FTP, HTTP, SMTP

Application

application specific needs

TCP, UDP

Transport

process to process data transfer

IP

Network

host to host data transfer across different network

Ethernet, WiFi

Link

data transfer between physically adjacent nodes

802.3 PHY

Physical

bit-by-bit or symbol-by-symbol delivery

# Outline

 I. TCP overview

# TCP: overview

RFCs: 793, 1122, 2018, 5681, 7323

## ▪ point-to-point:

- one sender, one receiver

## ▪ reliable, in-order byte stream:

- no “message boundaries”

## ▪ full duplex data:

- bi-directional data flow in same connection
- MSS: maximum segment size

## ▪ cumulative ACKs

- ACK N acks all packets till N-1 cumulatively
- ACK N means send me Seq N next

## ▪ timeouts

## ▪ pipelining:

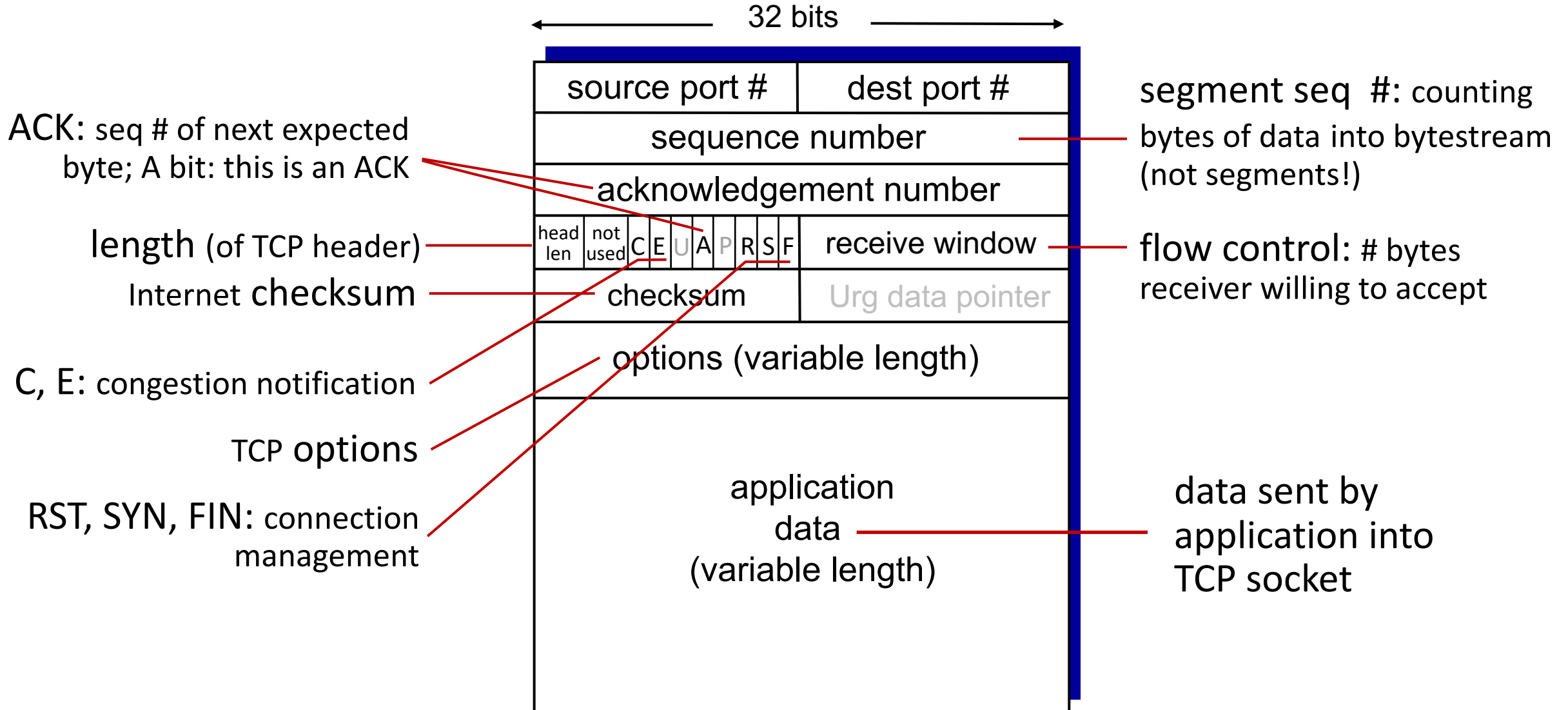
- TCP congestion and flow control set window size

## ▪ connection-oriented (handshake)

## ▪ flow controlled:

- sender will not overwhelm receiver

# TCP segment structure



# TCP sequence numbers, ACKs

## Sequence numbers:

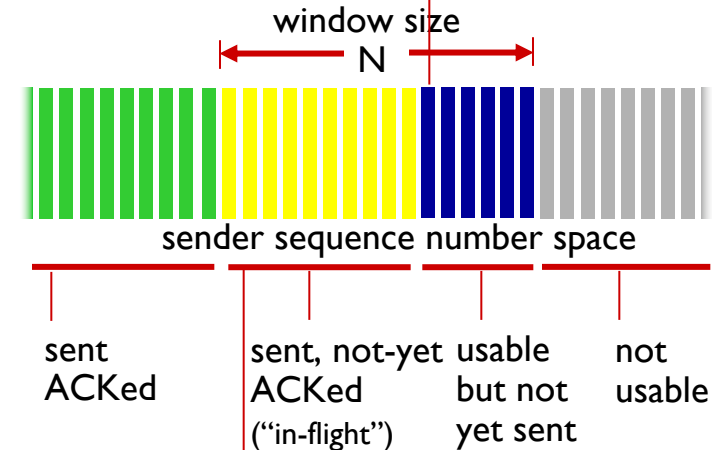
- byte stream “number” of first byte in segment’s data

## Acknowledgements:

- seq # of next byte expected from other side
- cumulative ACK

outgoing segment from sender

source port #	dest port #
sequence number	
acknowledgement number	
	rwnd
checksum	urg pointer

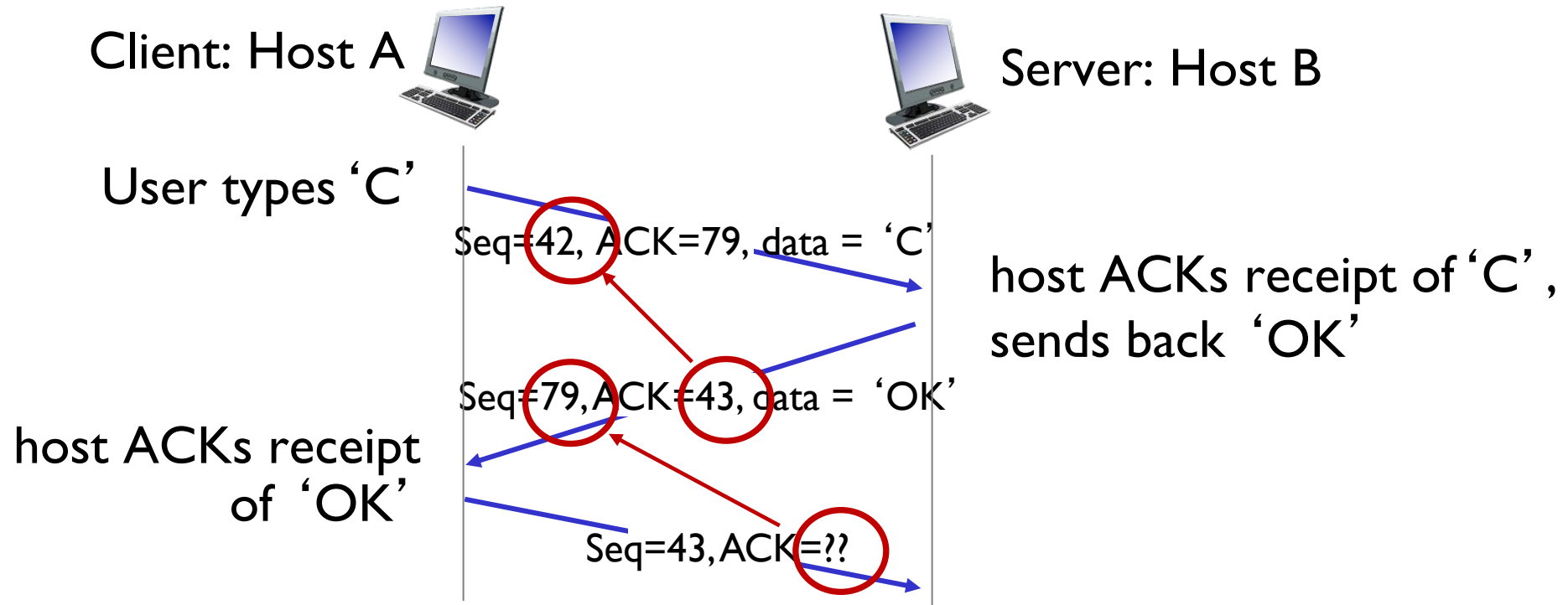


outgoing segment from receiver

source port #	dest port #
sequence number	
acknowledgement number	
	A
	rwnd
checksum	urg pointer

# TCP ACK $n$ means “please send me Seq $n$ ”

## simple telnet scenario

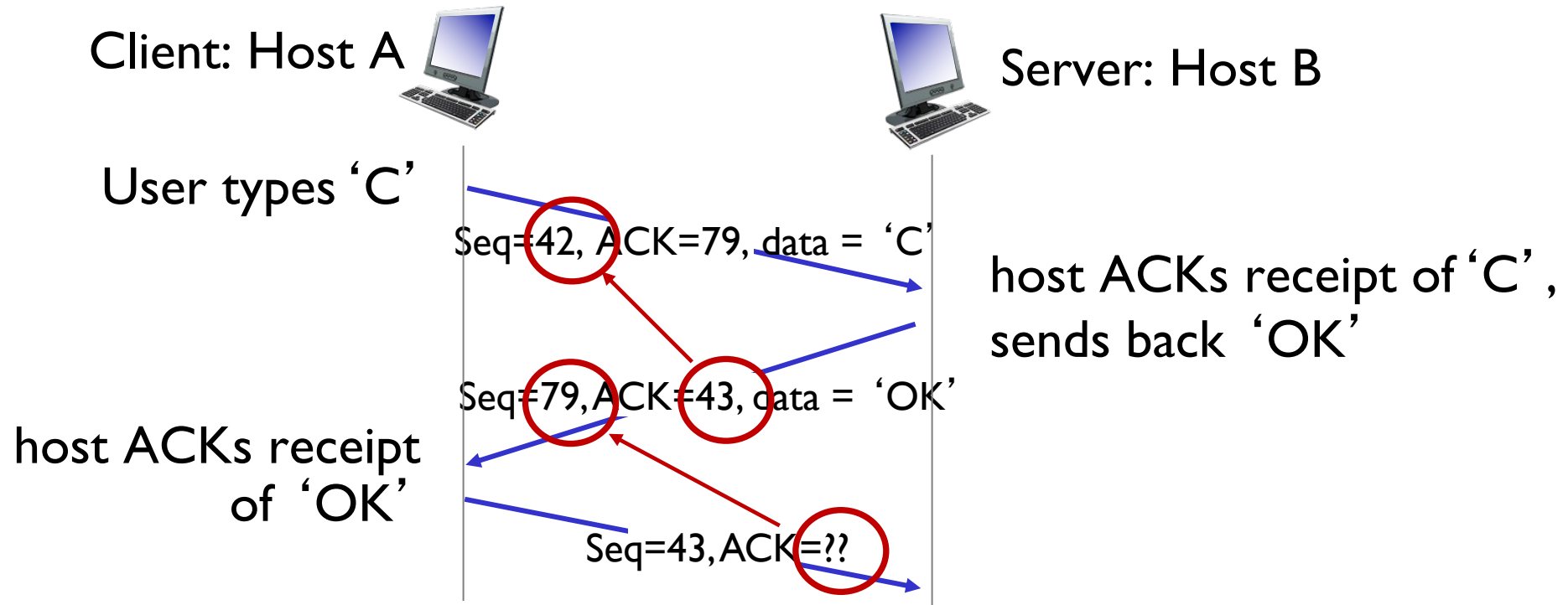


Seq number is calculated based on bytes sent

Num bytes sent is determined by network condition

# TCP ACKs can piggyback to DATA

## simple telnet scenario

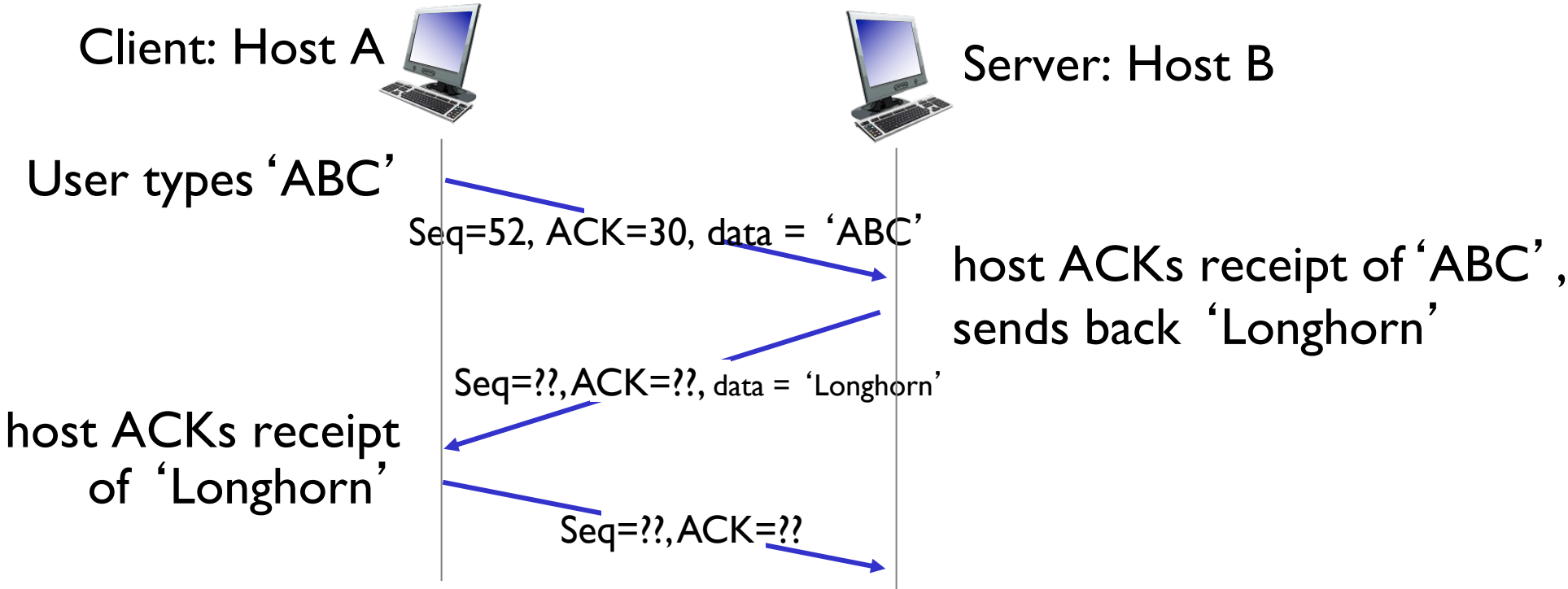


Which segments have the ACKs piggybacked to DATA?

Does the last segment have DATA? Why then seq no?

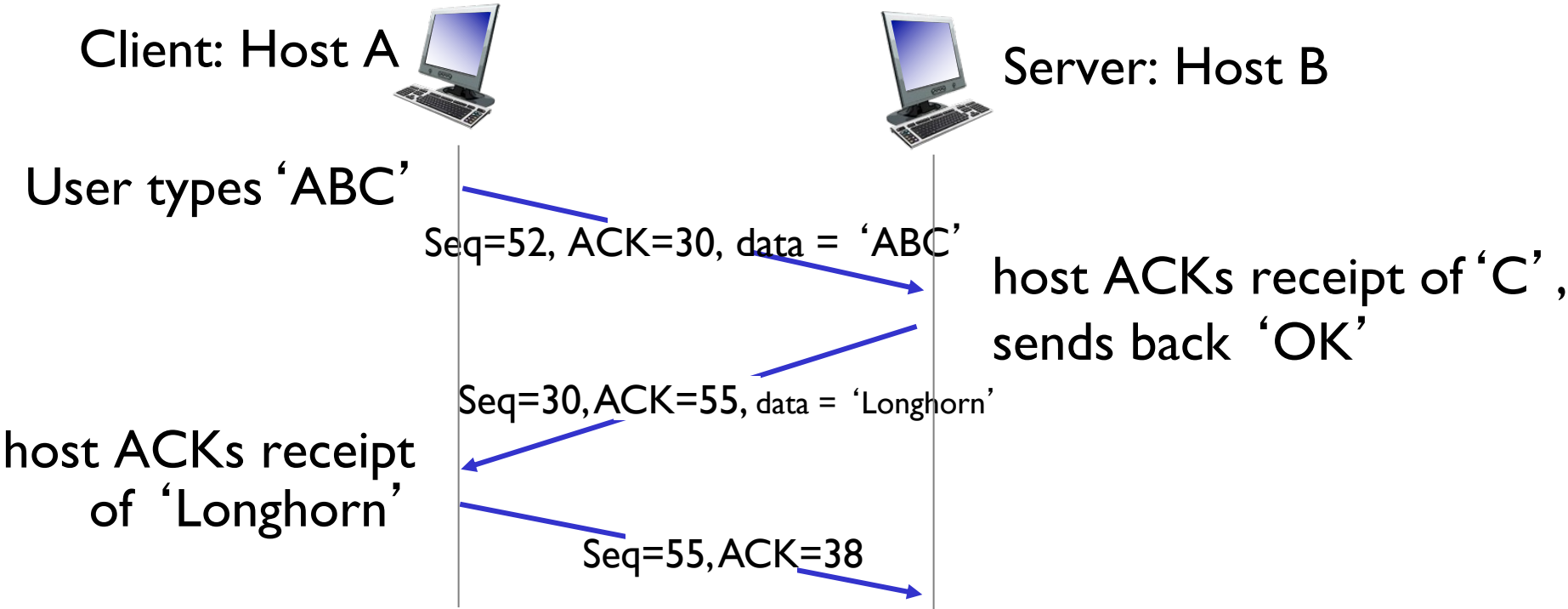
# Practice one more time!

## simple telnet scenario



# Practice one more time!

## simple telnet scenario



# Outline

1. TCP overview

 2. TCP timeout

# How to set TCP timeout value?

- What happens if timeout value is too short?
- What happens if timeout value is too long?
- We know it should be at least longer than... what?

# How to set TCP timeout value?

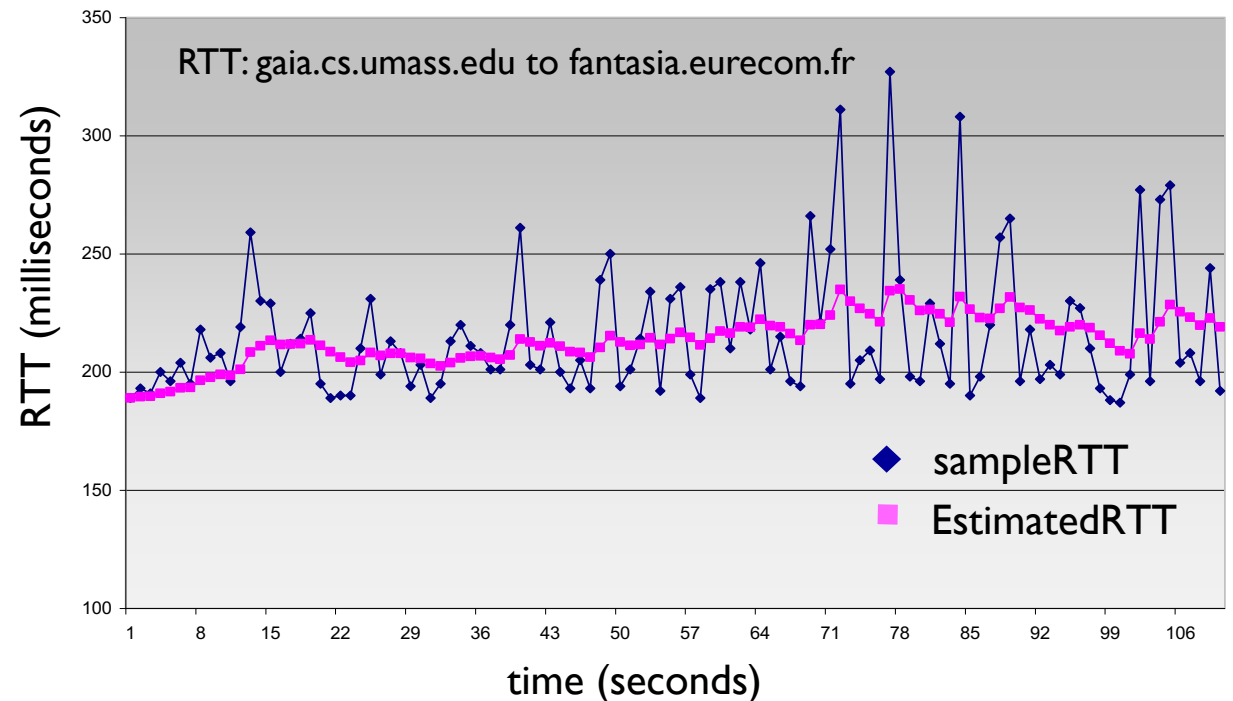
- **too short:** premature timeout, unnecessary retransmissions
- **too long:** slow reaction to segment loss
- It should be at least longer than RTT but RTT varies!
- TCP maintains timer for its **oldest unACKed segment**

TCP uses EWMA of Sample RTT plus safety margin

# Estimate RTT uses EWMA to smooth out

$$\text{EstimatedRTT}_n = (1 - \alpha) * \text{EstimatedRTT}_{n-1} + \alpha * \text{SampleRTT}_n$$

- exponential weighted moving average (EWMA)
- SampleRTT: measured time from segment transmission until ACK receipt
- influence of past sample decreases exponentially fast
- typical value:  $\alpha = 0.125$



# In addition, safety margin is added

- timeout interval: EstimatedRTT plus “safety margin”
  - large variation in EstimatedRTT: want a larger safety margin

$$\text{TimeoutInterval} = \text{EstimatedRTT} + 4 * \text{DevRTT}$$



↑  
estimated RTT


↑  
“safety margin”

- DevRTT: EWMA of SampleRTT deviation from EstimatedRTT:

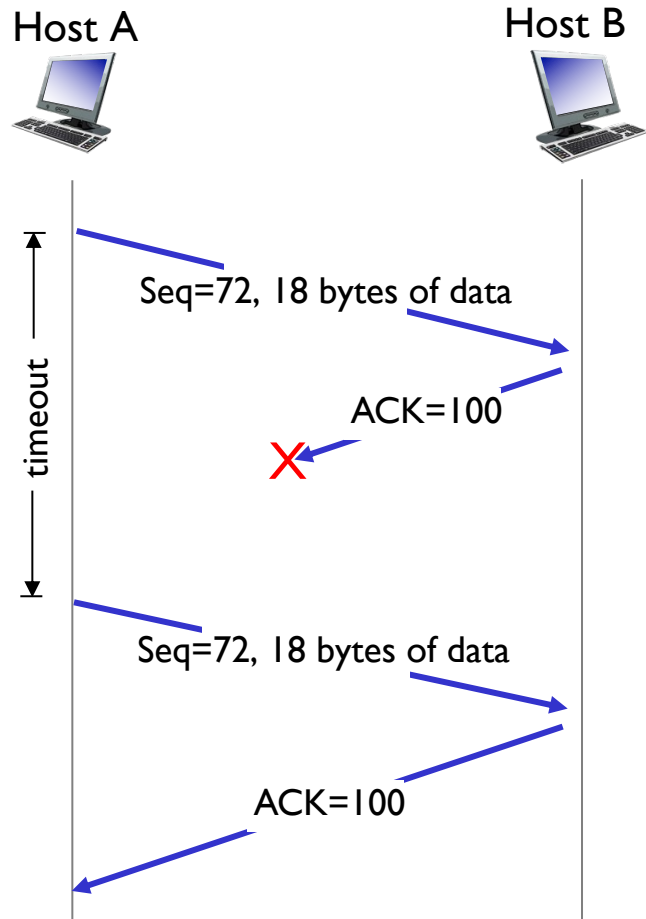
$$\text{DevRTT}_n = (1 - \beta) * \text{DevRTT}_{n-1} + \beta * |\text{SampleRTT}_n - \text{EstimatedRTT}_n|$$

(typically,  $\beta = 0.25$ )

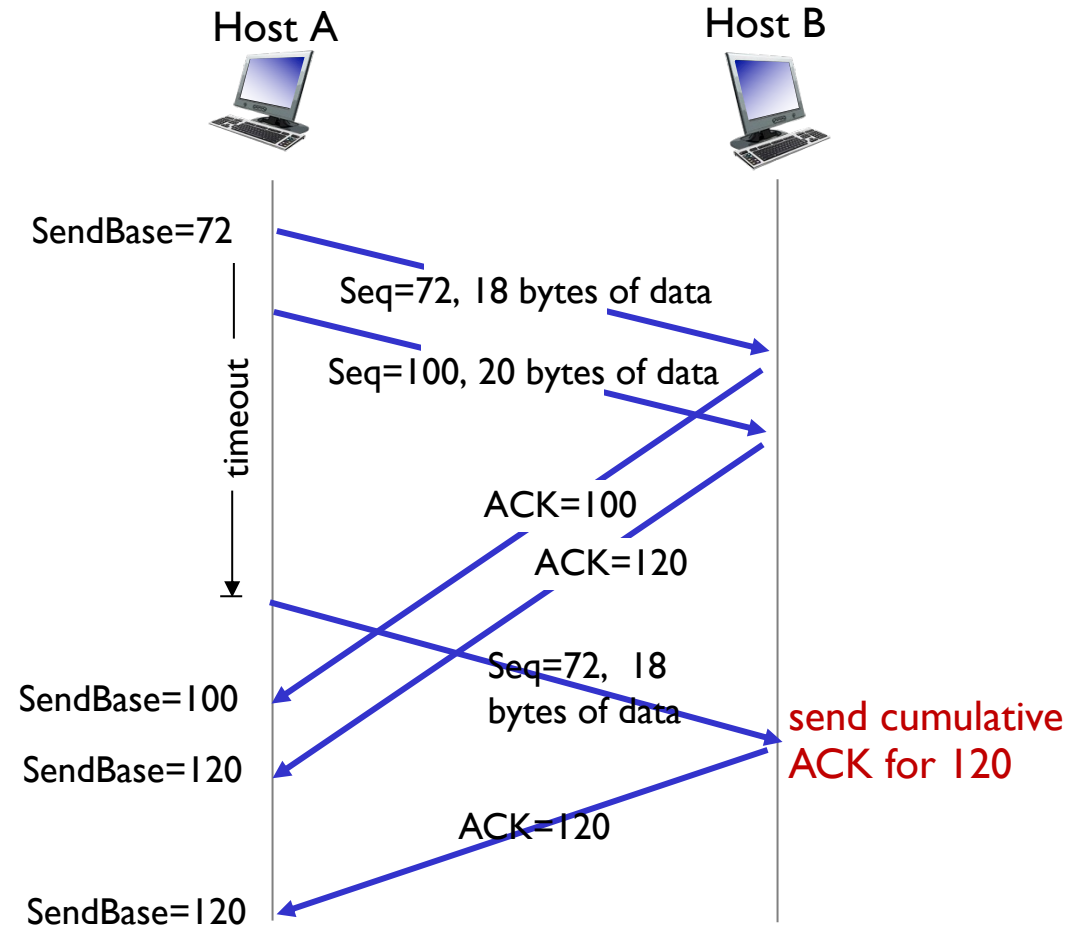
# Outline

1. TCP overview
2. TCP timeout
-  3. TCP retransmissions

# TCP: retransmission scenarios

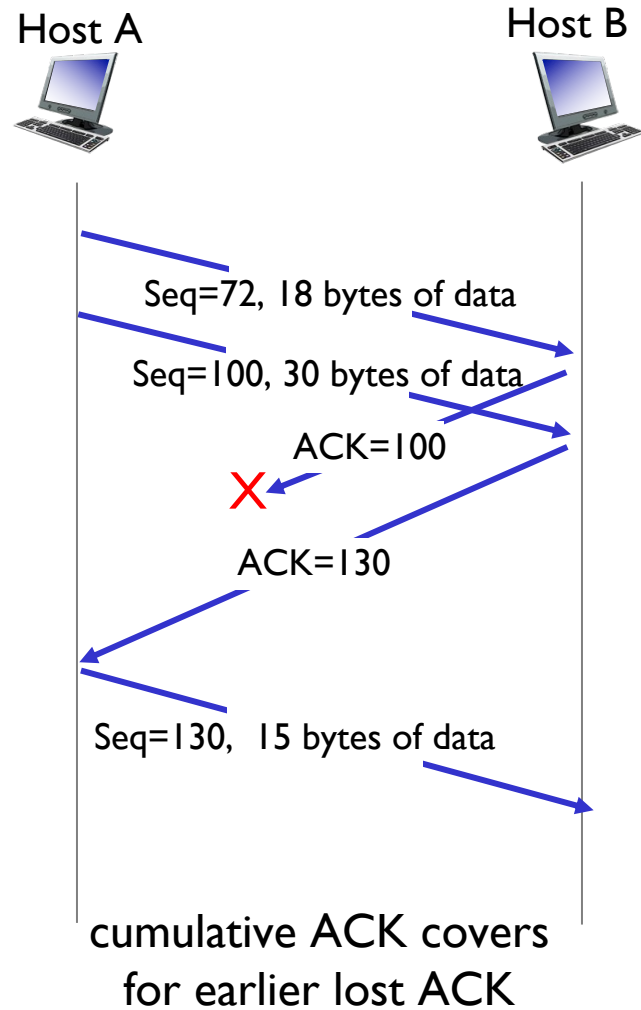


lost ACK scenario



premature timeout

# TCP: retransmission scenarios






# Is it a good idea to retransmit as soon as possible?

- TCP assumes packet is lost upon timeout
- TCP assumes the packet is lost due to congestion

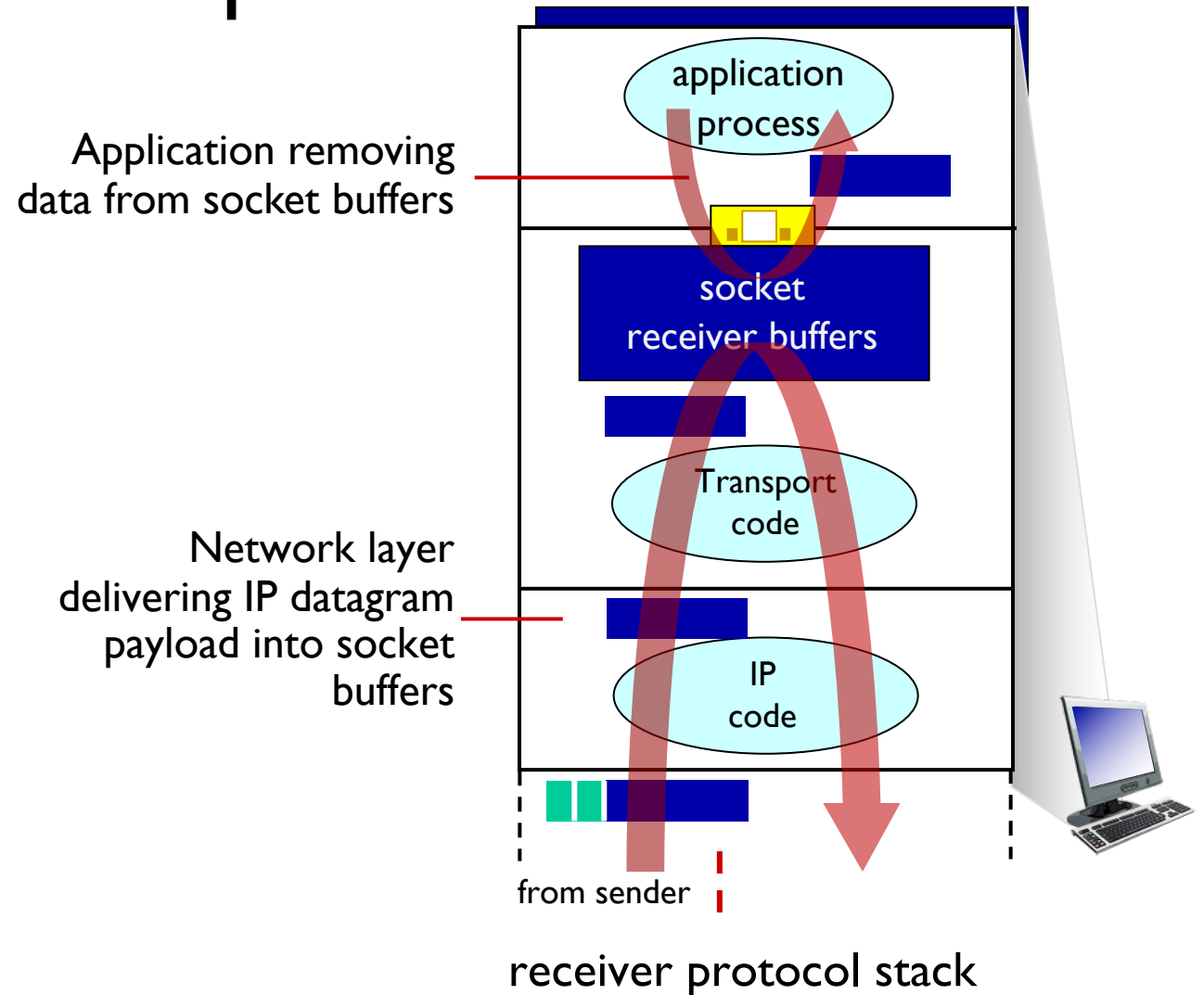
Doubles the timeout interval each time TCP retransmits upon timeout!

# Outline

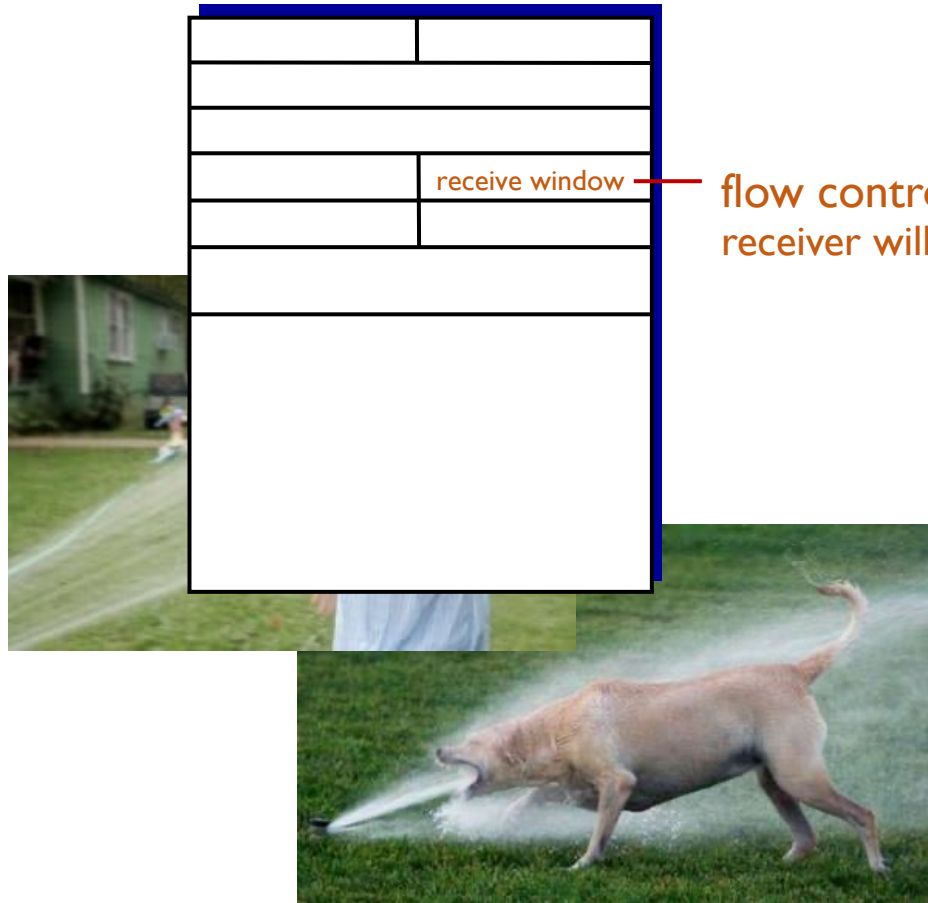
1. TCP overview
2. TCP timeout
3. TCP interesting scenarios
-  4. **TCP flow control**

# Loss happens if network delivers faster than what application layer can process

Loss was happening in the socket buffer of the receiver!



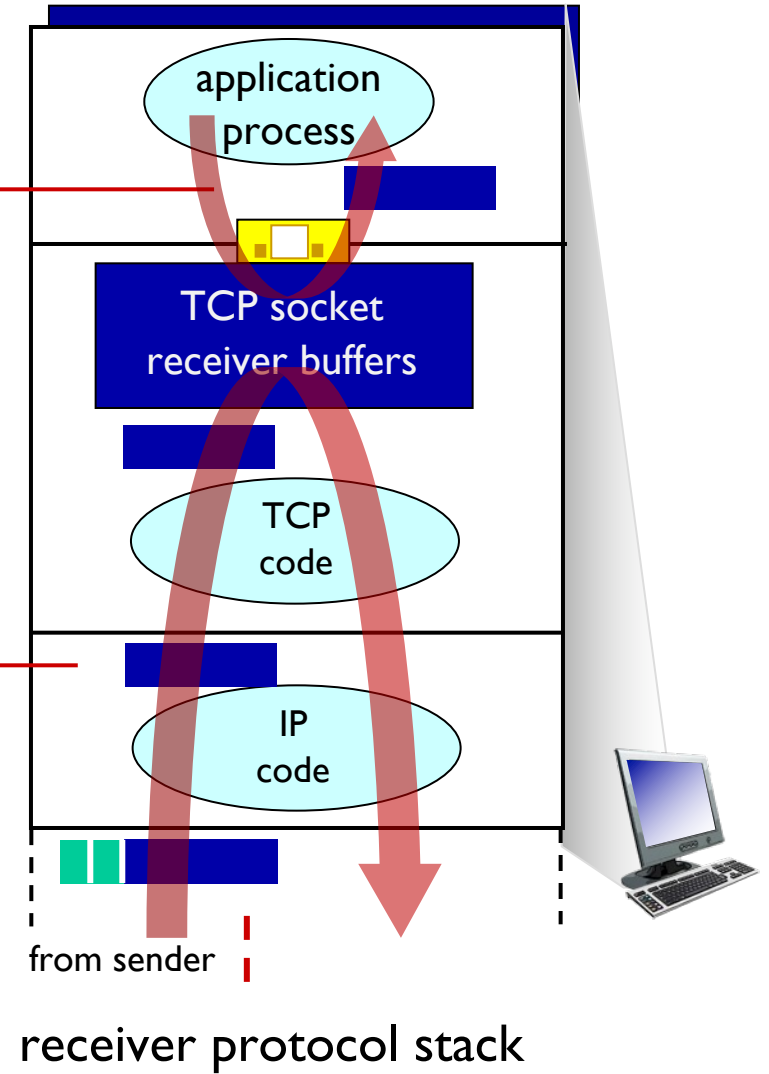
# TCP flow control ensures NOT to overflow receiver socket buffer



Application removing data from TCP socket buffers

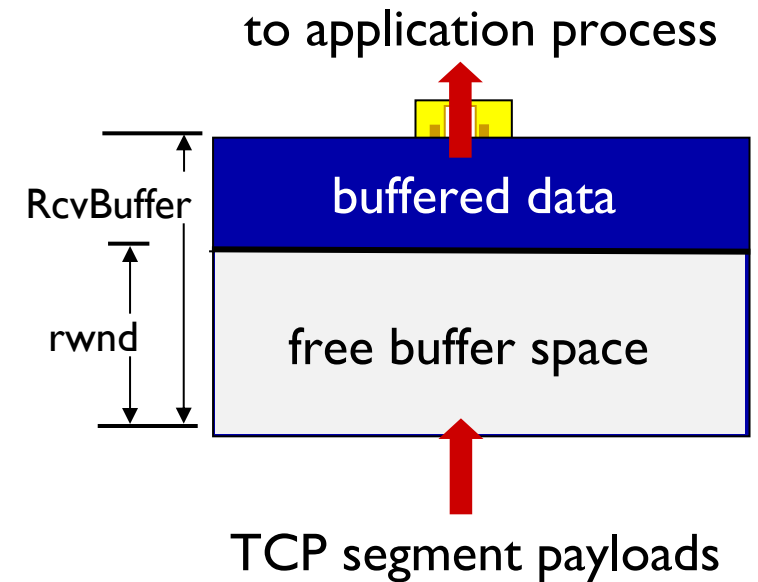
flow control: # bytes receiver willing to accept

Network layer delivering IP datagram payload into TCP socket buffers



# TCP sender limits in-flight packets smaller than rwnd

- TCP receiver “advertises” free buffer space in rwnd field in TCP header
  - RcvBuffer size set via socket options (default 4096 bytes)



TCP receiver-side buffering

Guarantees receiver buffer will not overflow!

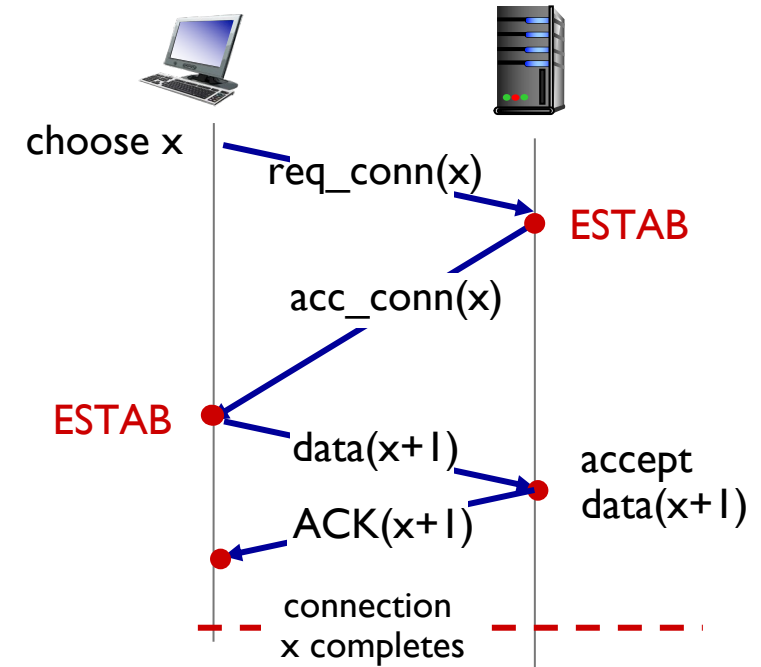
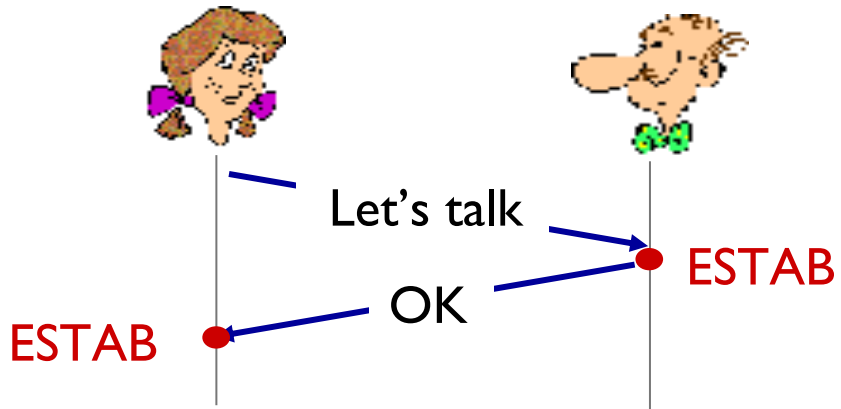
# Outline

1. TCP overview
2. TCP timeout
3. TCP interesting scenarios
4. TCP flow control
-  5. **TCP connection management**

# TCP has “handshake” prior to actual data exchange

- agree to establish connection
- agree on connection parameters (e.g., starting seq #s, rwnds)

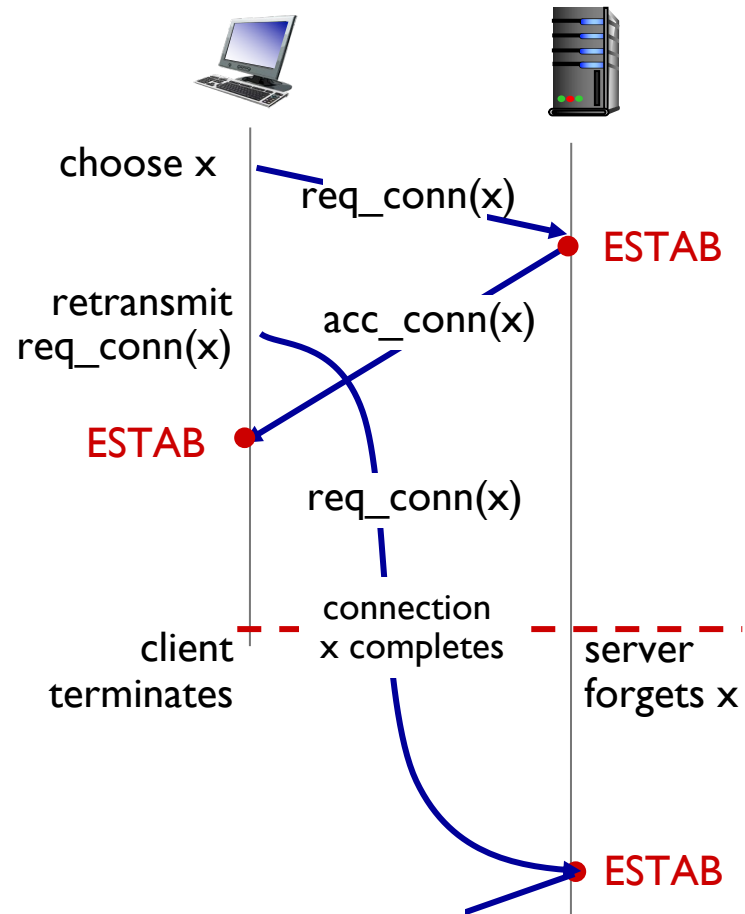
# We could use 2-way handshake




No problem!

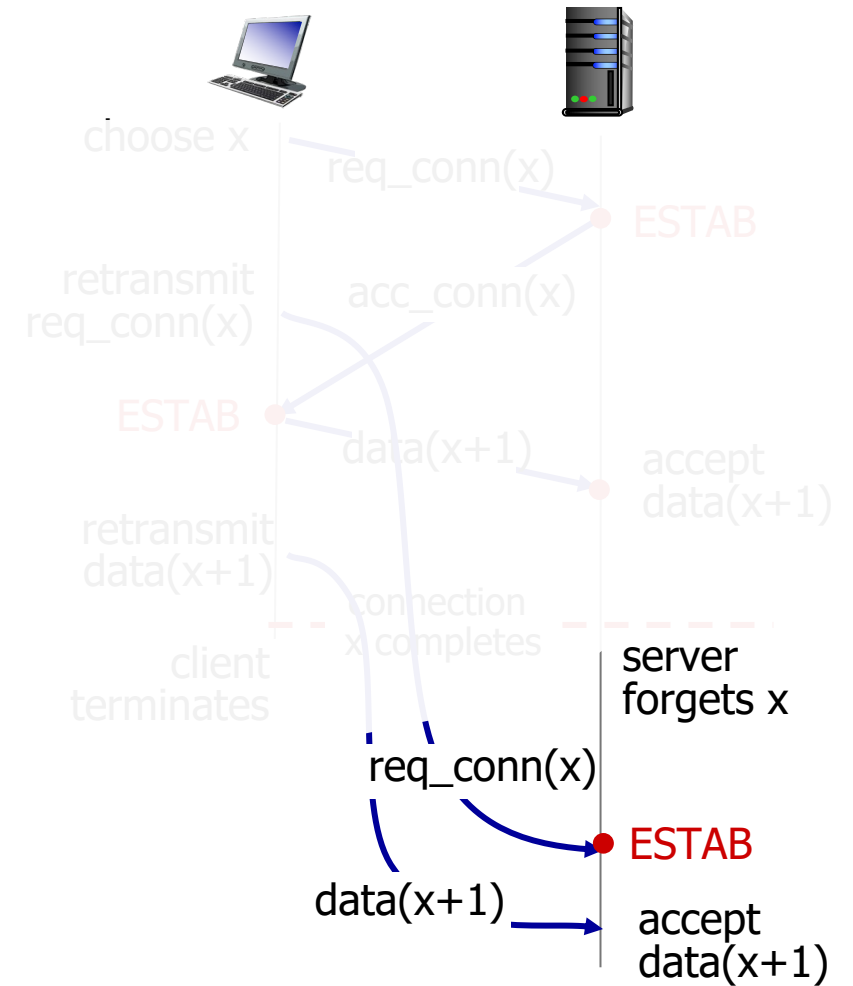



# 2-way handshake is not enough!



 Problem: half open connection! (no client)

# 2-way handshake is not enough!



 Problem: dup data accepted!

# TCP 3-way handshake

## Client state

```
clientSocket = socket(AF_INET, SOCK_STREAM)
```

LISTEN

```
clientSocket.connect((serverName,serverPort))
```

SYNSENT

ESTAB

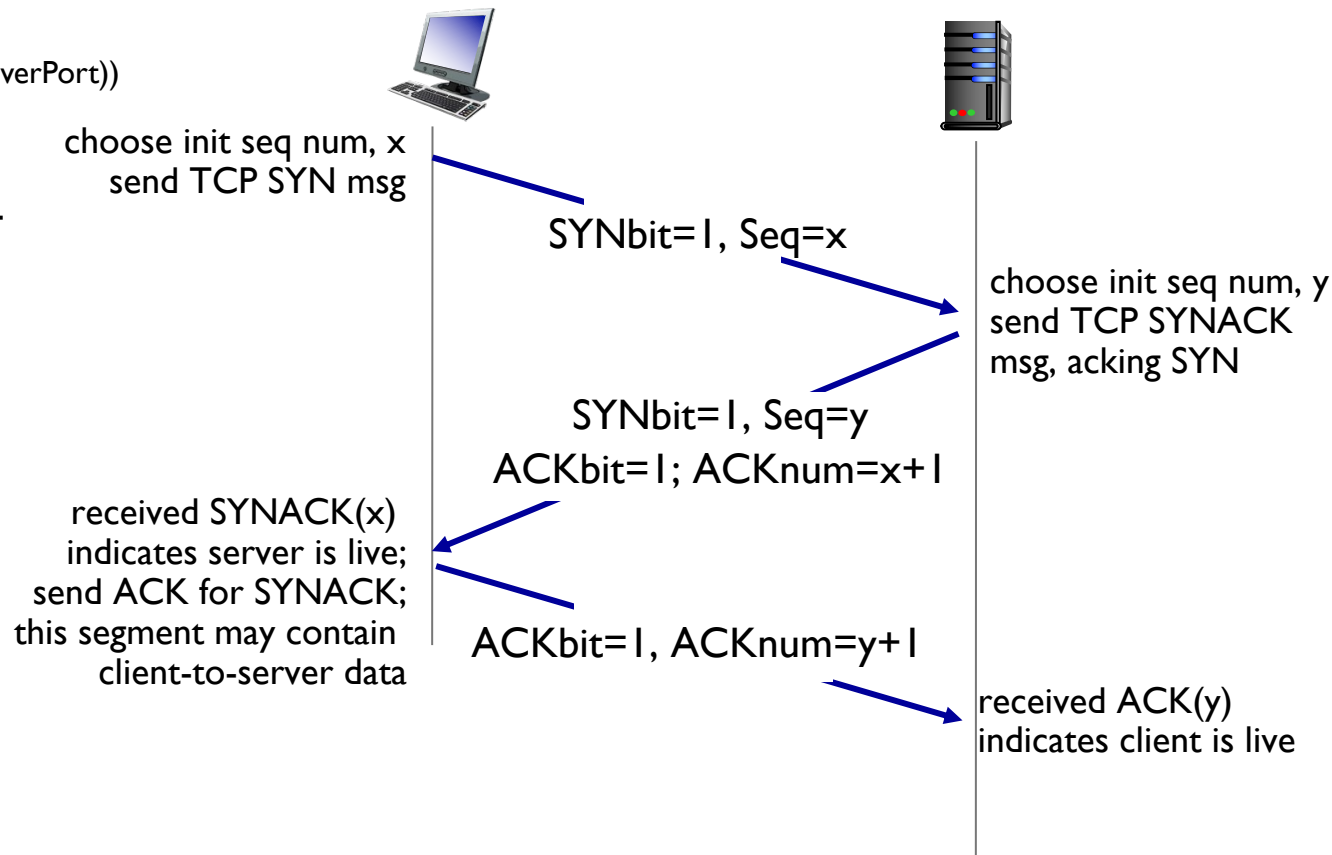
## Server state

```
serverSocket = socket(AF_INET,SOCK_STREAM)  
serverSocket.bind(("",serverPort))  
serverSocket.listen(1)  
connectionSocket, addr = serverSocket.accept()
```

LISTEN

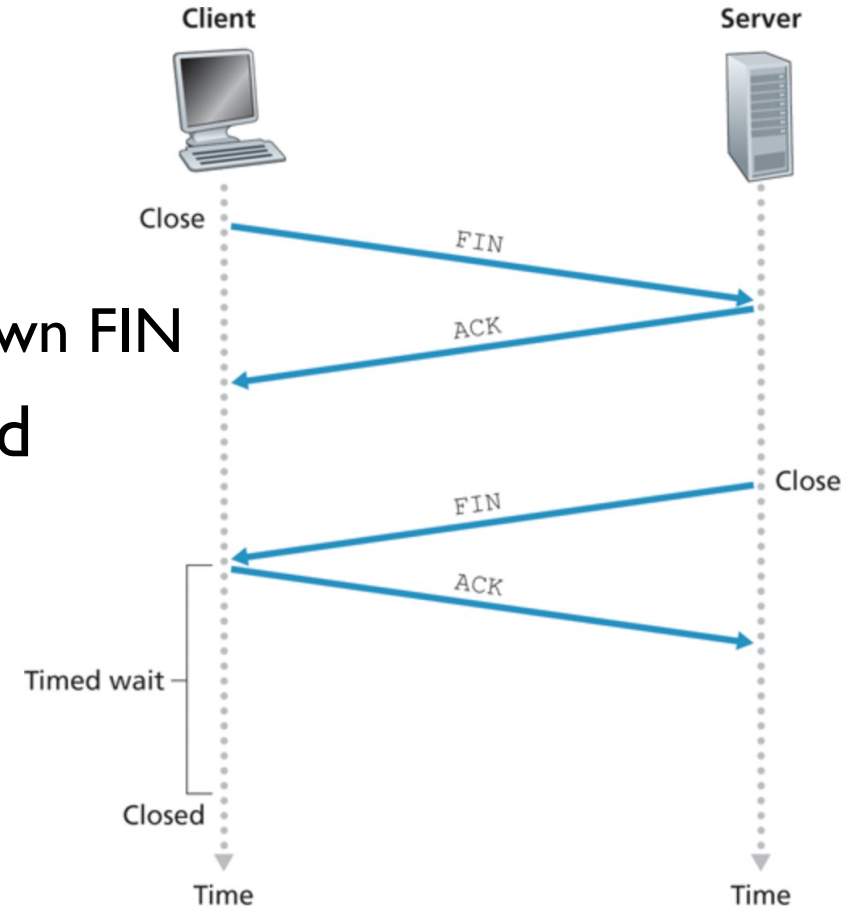
SYN RCVD

ESTAB



# Closing a TCP connection

- Send TCP segment with FIN bit = 1
- respond to received FIN with ACK
  - on receiving FIN, ACK can be combined with own FIN
- simultaneous FIN exchanges can be handled



# Backup slides

# Acknowledgements

Slides are adopted from Kurose' Computer Networking Slides