Chapter 3 Transport Layer

Yaxiong Xie

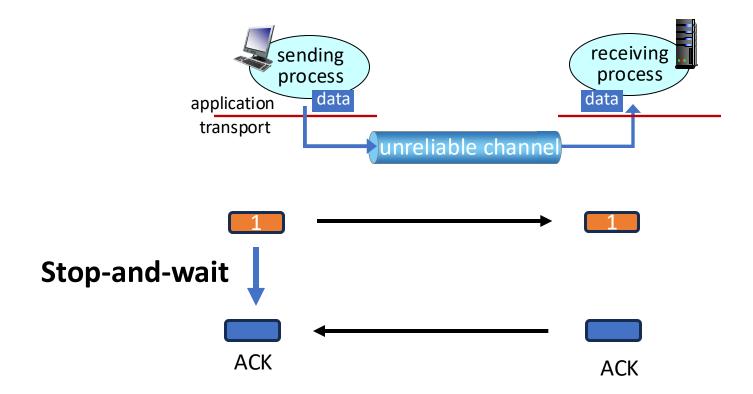
Department of Computer Science and Engineering University at Buffalo, SUNY

James F. Kurose | Keith W. Ross COMPUTER A TOP-DOWN APPROACH P

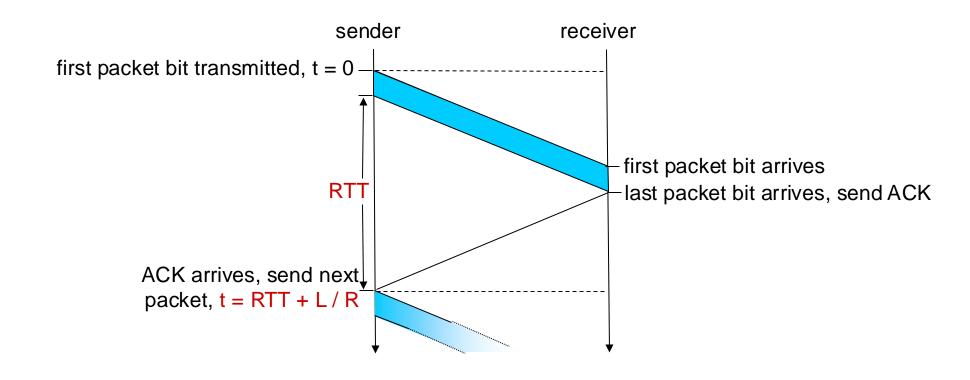
Computer Networking: A Top-Down Approach

8th edition Jim Kurose, Keith Ross Pearson, 2020

Stop-and-wait operation



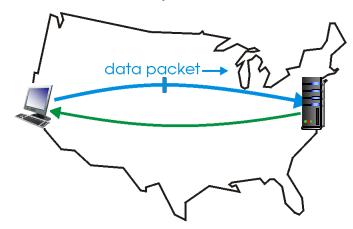
Stop-and-wait operation



Pipelined protocols operation

pipelining: sender allows multiple, "in-flight", yet-to-be-acknowledged packets

- range of sequence numbers must be increased
- buffering at sender and/or receiver

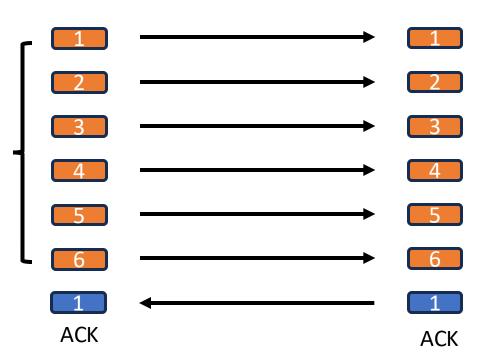


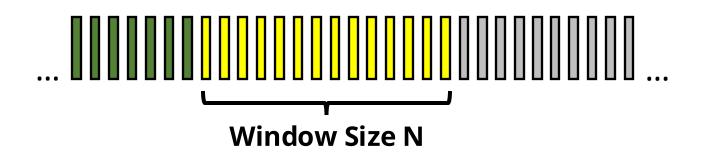
(a) a stop-and-wait protocol in operation

Pipelining



How many packets shall we sent before receiving the ACK?

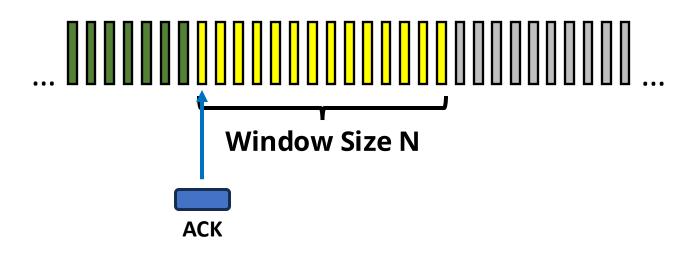




sent and ACKed

Not yet sent

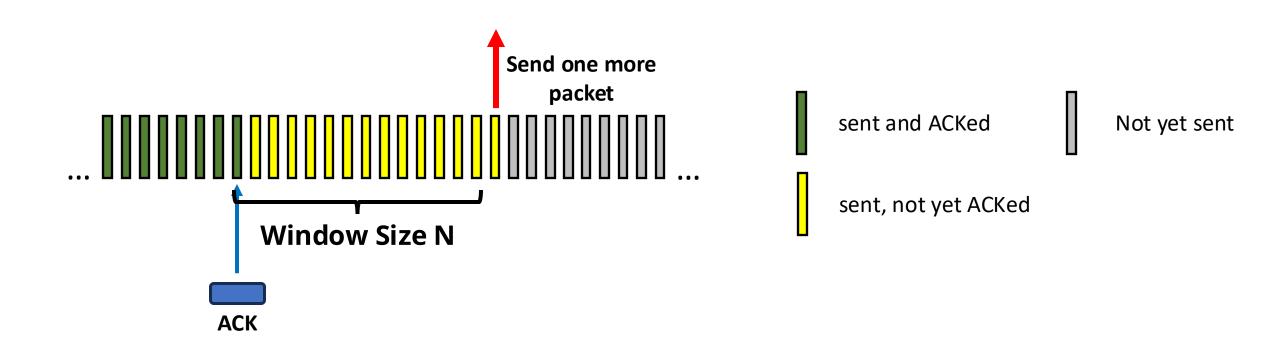
sent, not yet ACKed

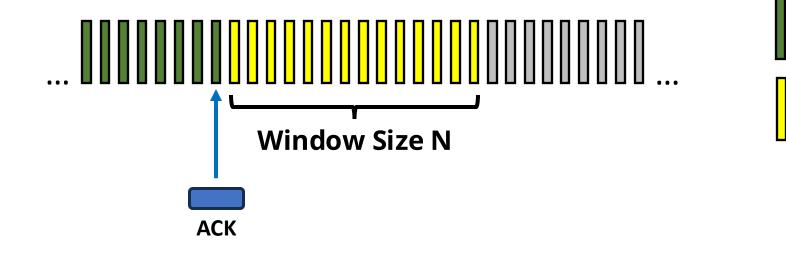


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Not yet sent

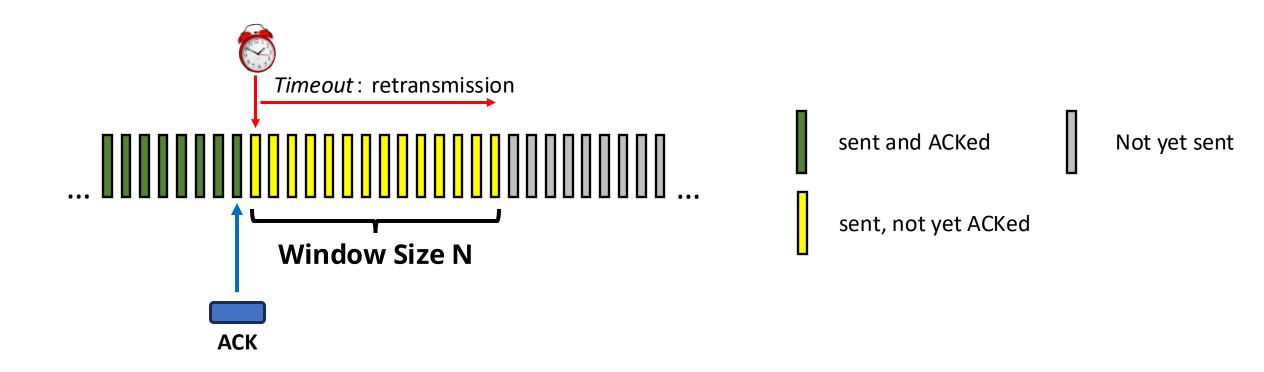




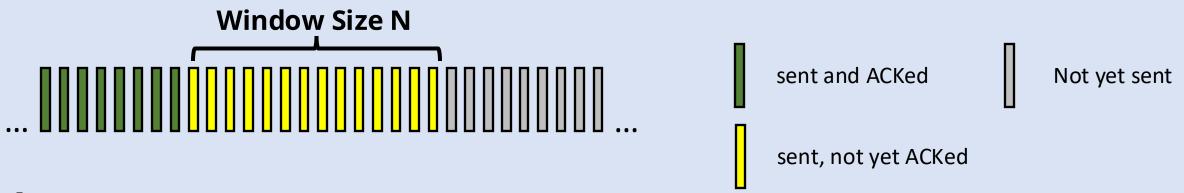
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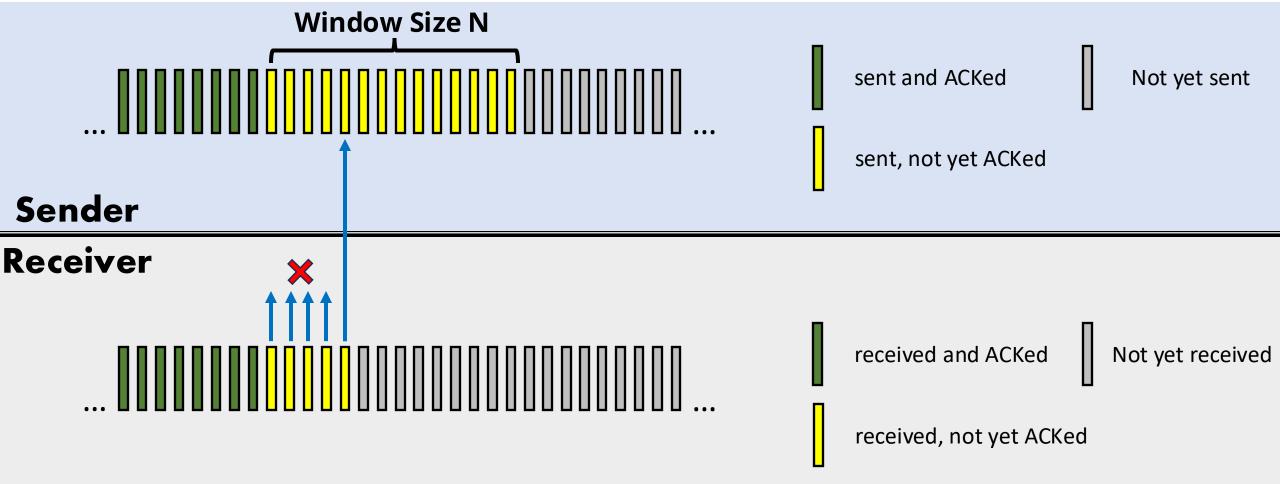
- Timer for oldest in-flight packet
- Timeout(n): retransmit packet n and all higher seq # packets in window

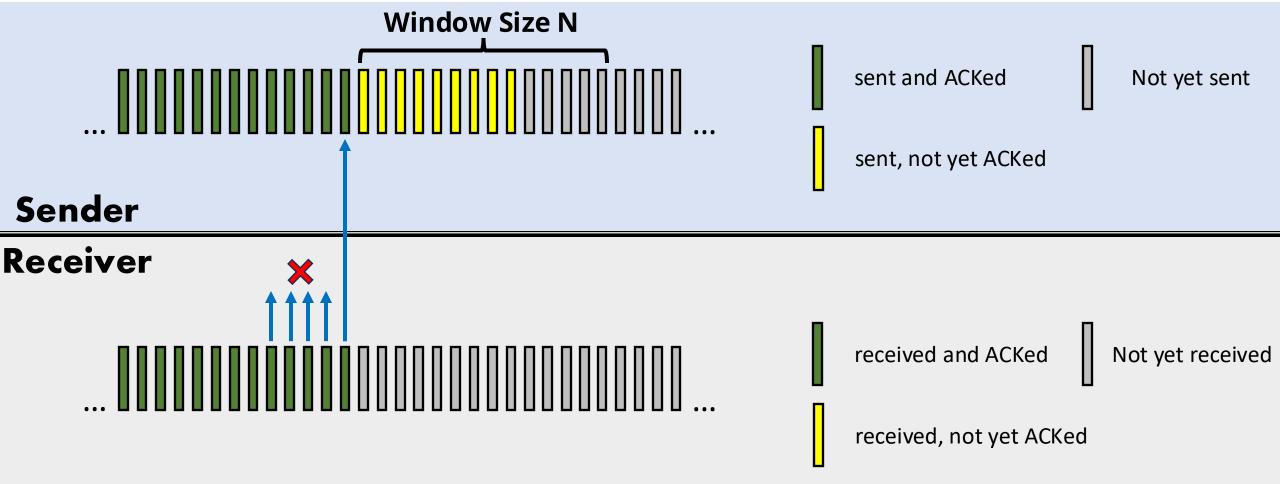


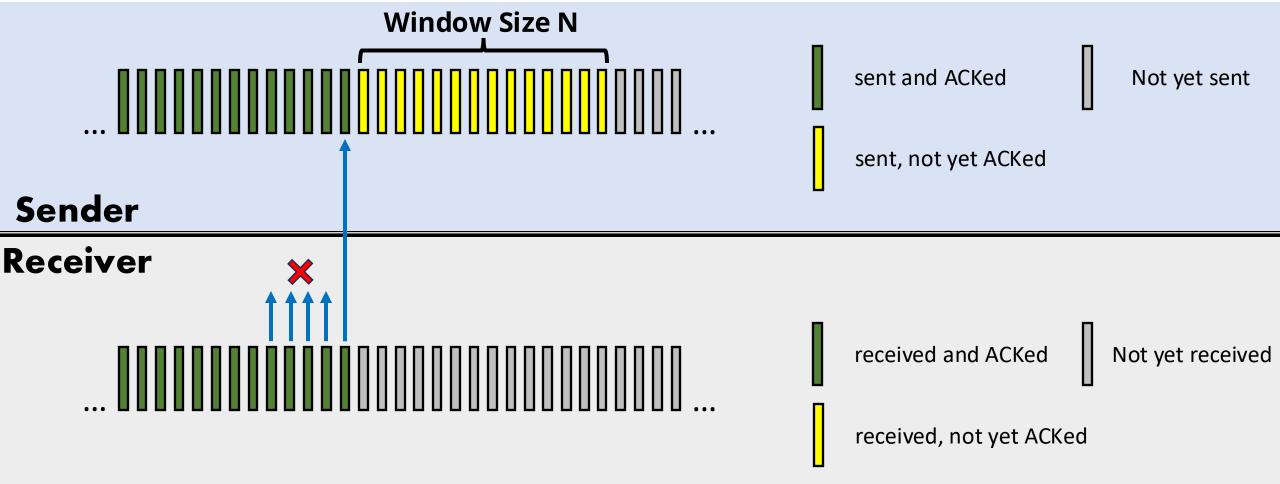
Sender

Receiver

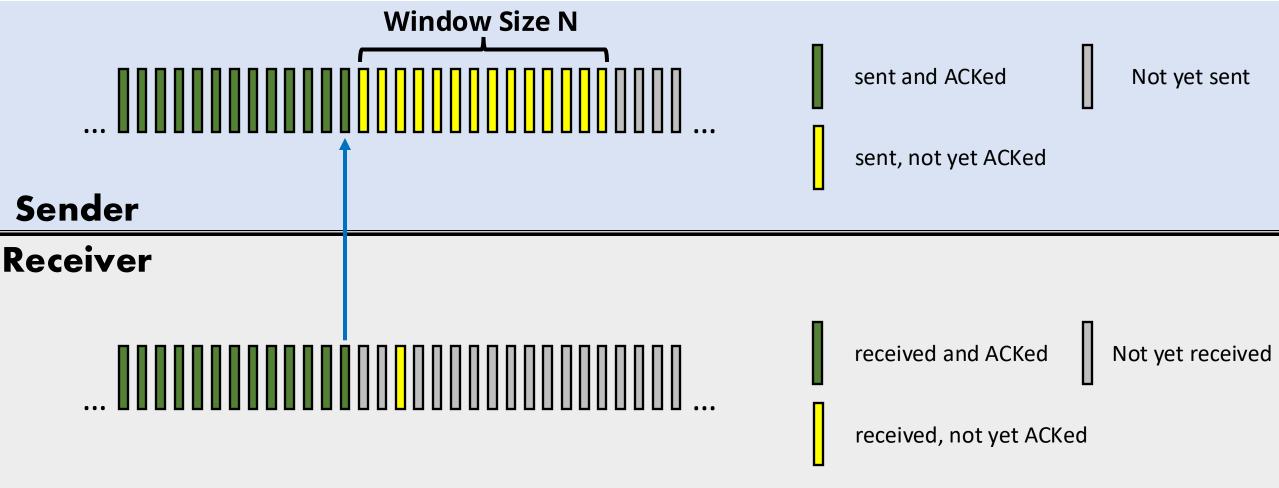






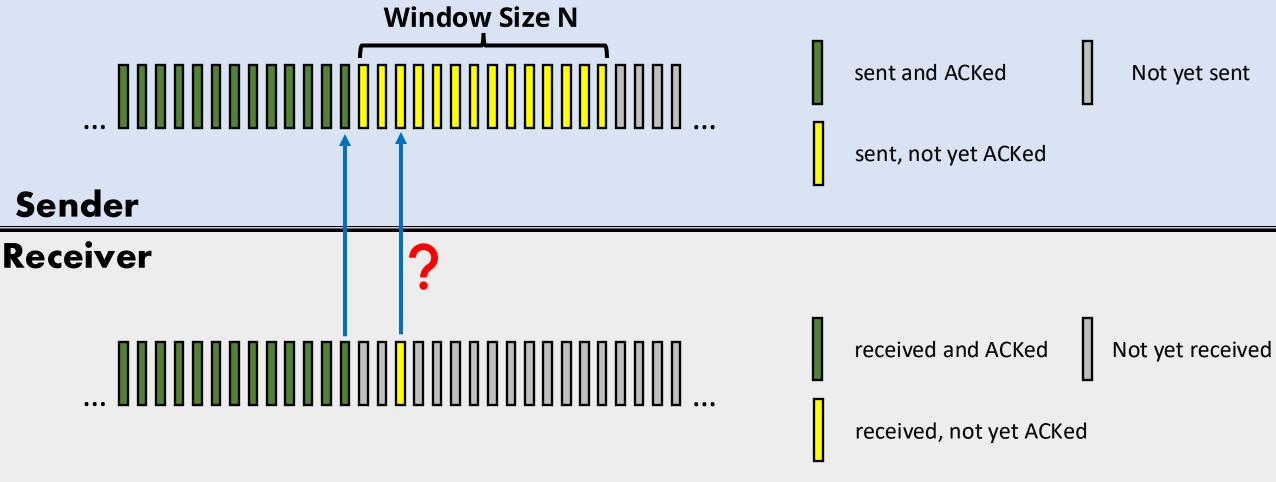


- cumulative ACK: ACK(n): ACKs all packets up to, including seq # n
 - on receiving ACK(n): move window forward to begin at n+1



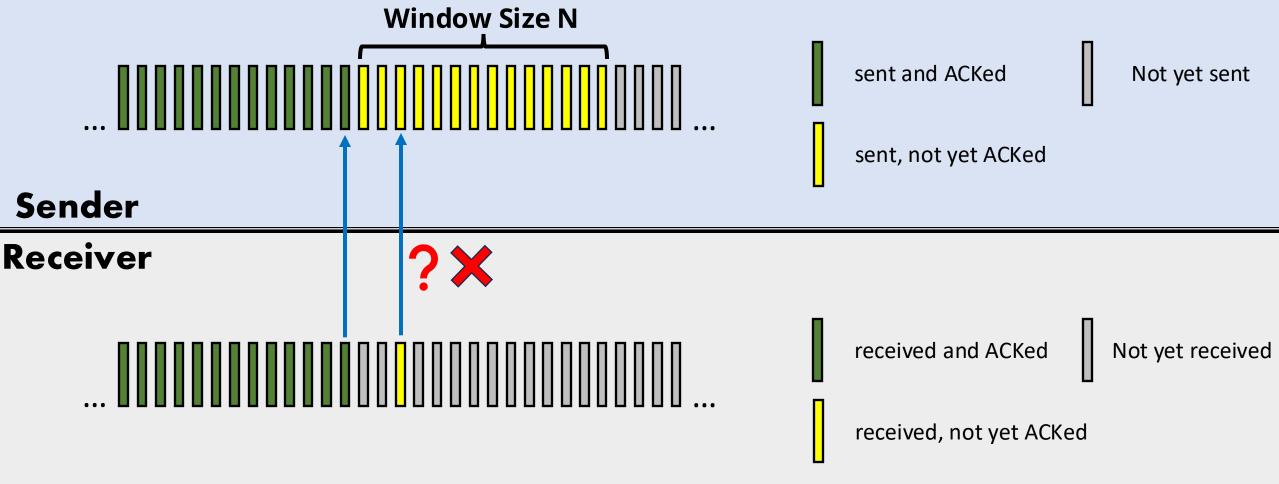
Two Question:

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- 2. how shall we deal with the out-of-order packets we received?

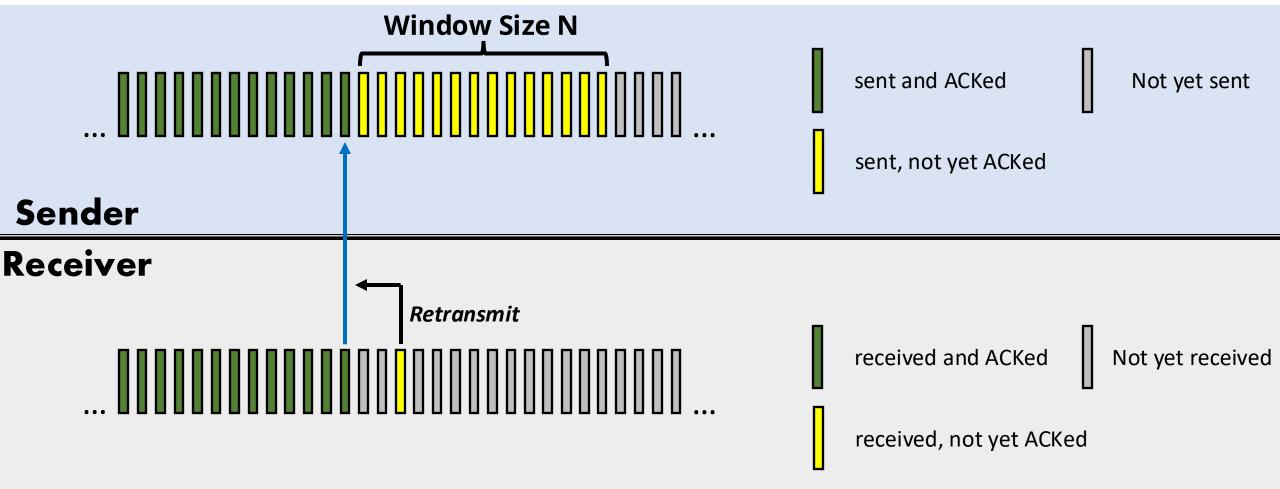


Two Question:

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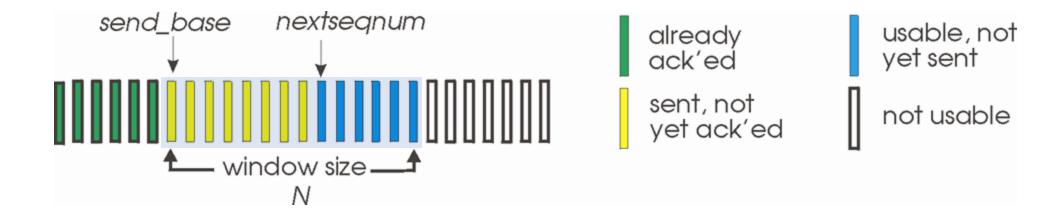
- ACK-only: always send ACK for correctly-received packet so far, with highest in-order seq #
 - may generate duplicate ACKs



- on receipt of out-of-order packet:
 - can discard (don't buffer) or buffer: an implementation decision
 - re-ACK pkt with highest in-order seq #

Go-Back-N: sender

- sender: "window" of up to N, consecutive transmitted but unACKed pkts
 - k-bit seq # in pkt header

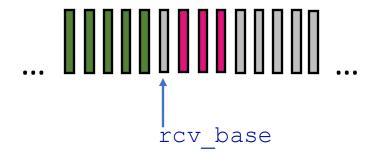


- cumulative ACK: ACK(n): ACKs all packets up to, including seq # n
 - on receiving ACK(n): move window forward to begin at n+1
- timer for oldest in-flight packet
- timeout(n): retransmit packet n and all higher seq # packets in window

Go-Back-N: receiver

- ACK-only: always send ACK for correctly-received packet so far, with highest in-order seq #
 - may generate duplicate ACKs
 - need only remember rcv base
 - on receipt of out-of-order packet:
 - can discard (don't buffer) or buffer: an implementation decision
 - re-ACK pkt with highest in-order seq #

Receiver view of sequence number space:

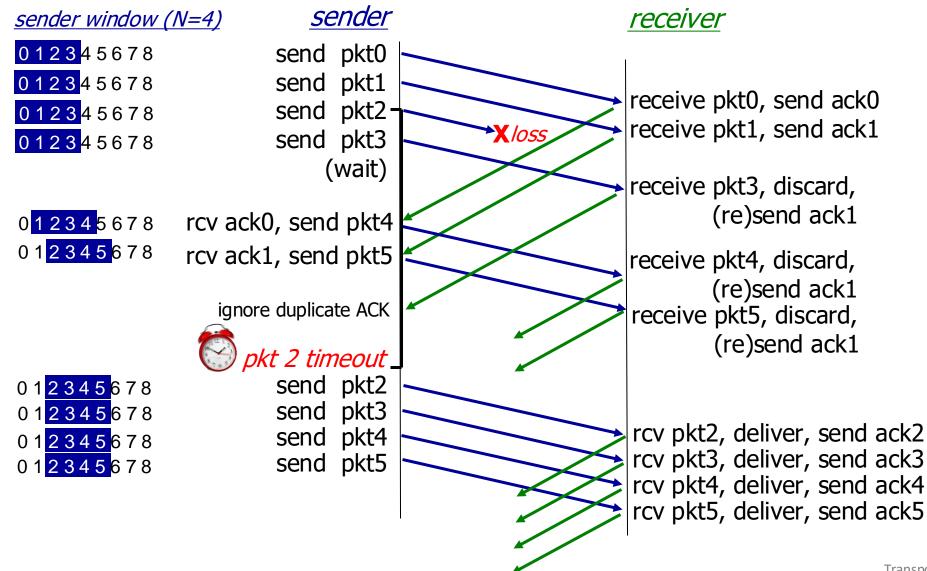


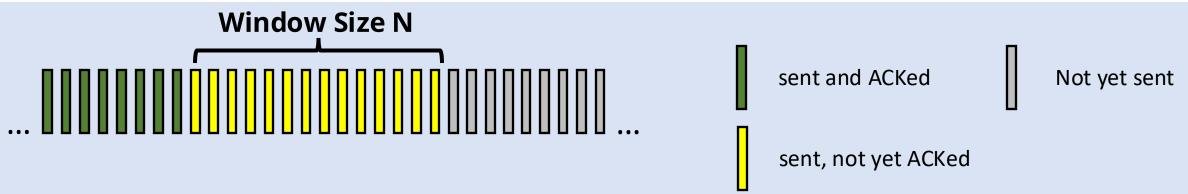
received and ACKed

Out-of-order: received but not ACKed

Not received

Go-Back-N in action

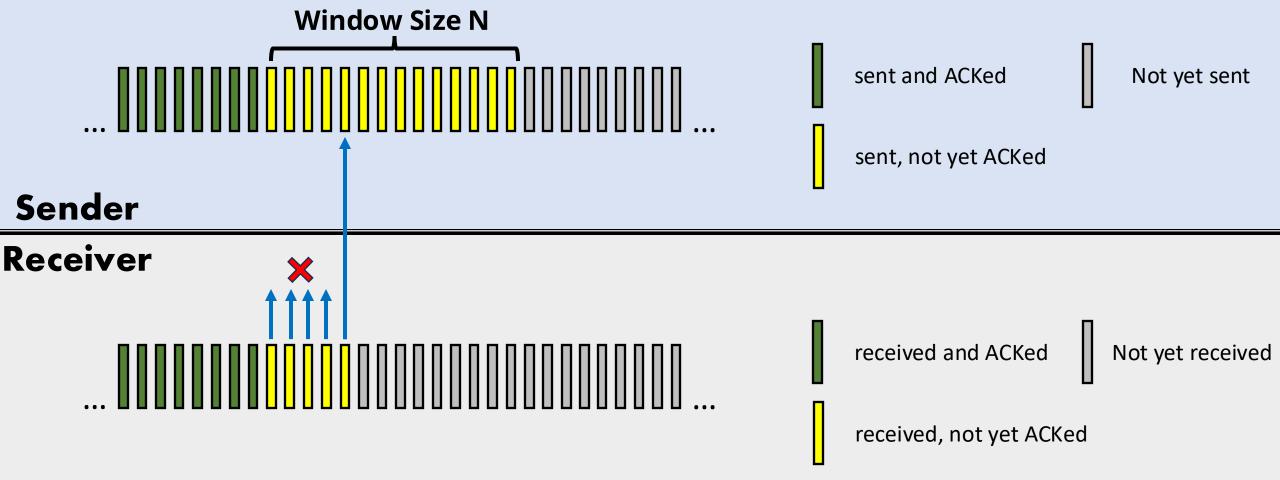


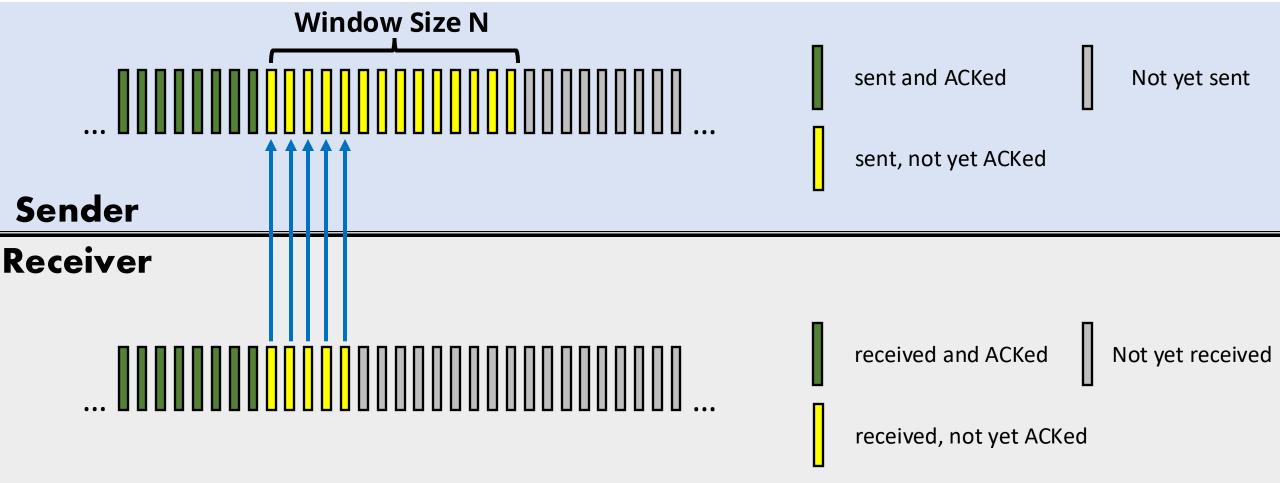


Sender

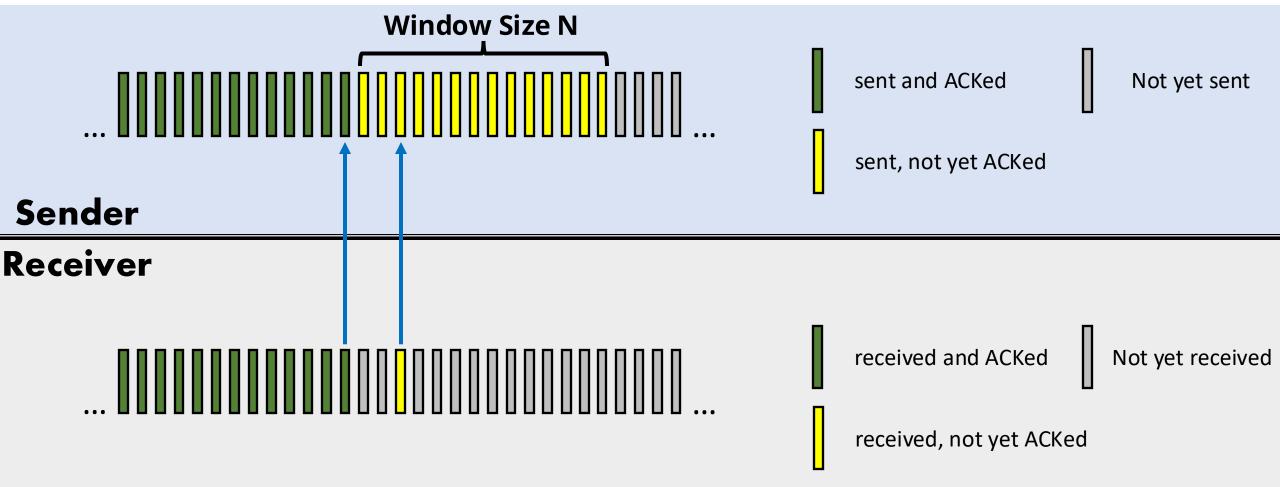
Receiver

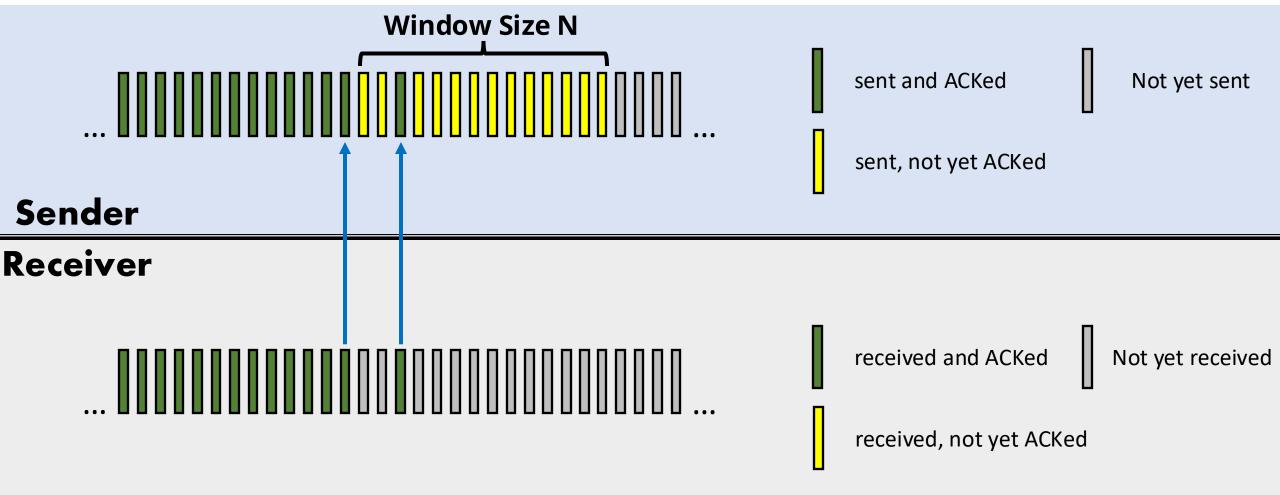




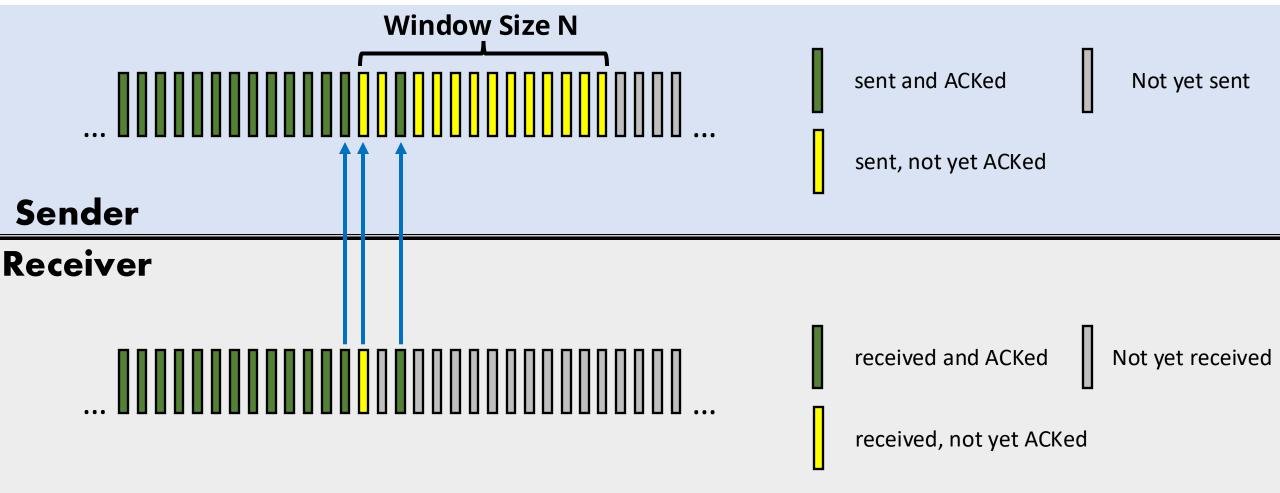




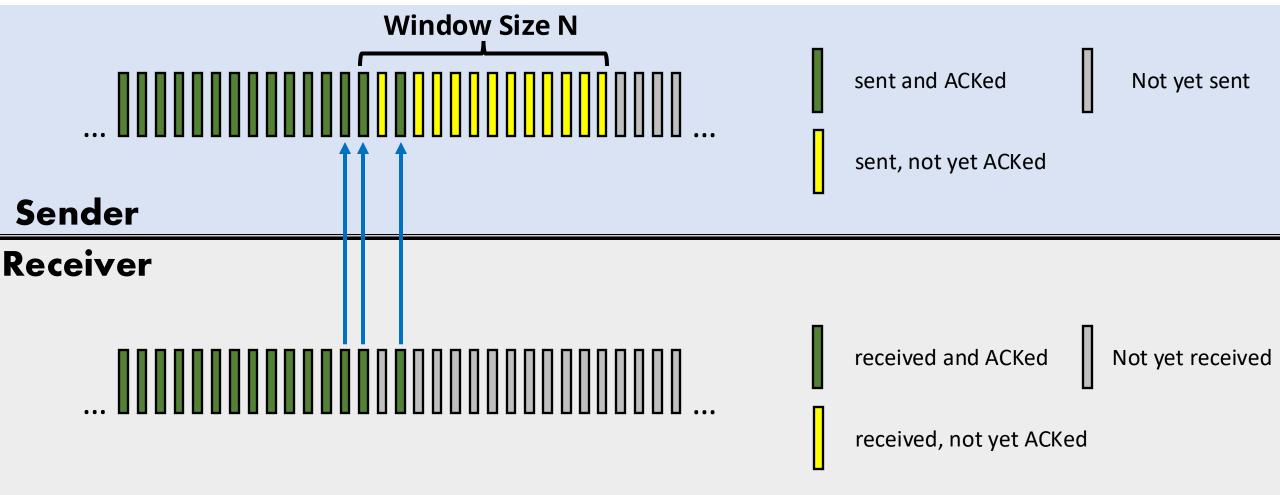




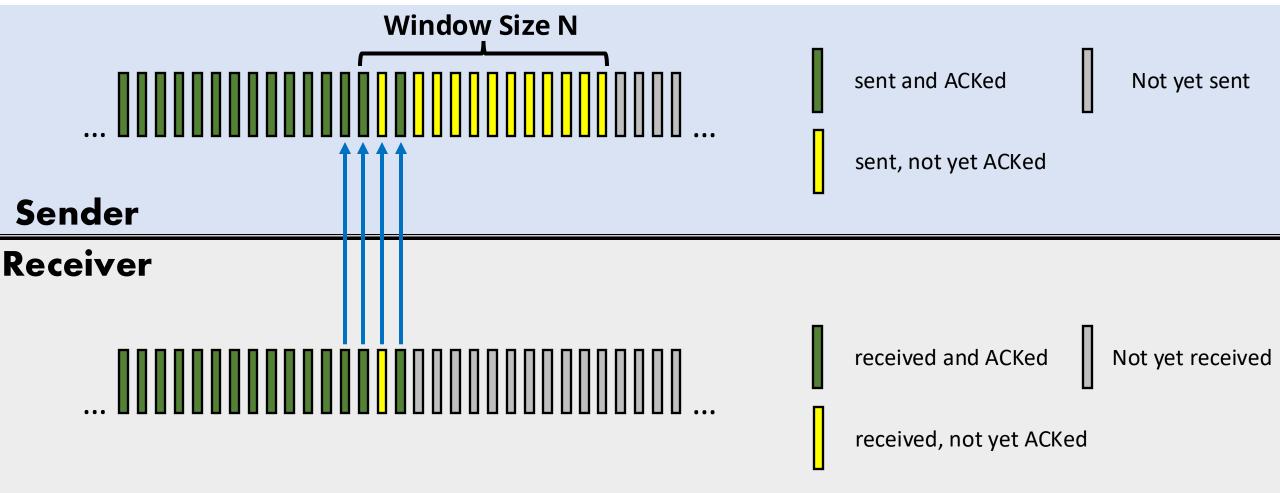
- receiver individually ACKs all correctly received packets
 - buffers packets, as needed, for in-order delivery to upper layer



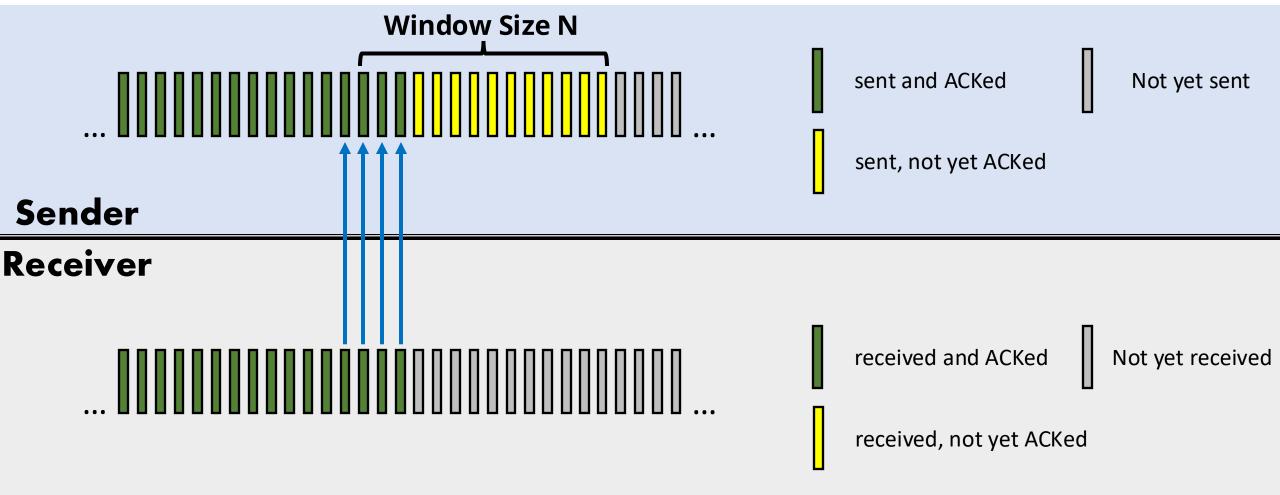
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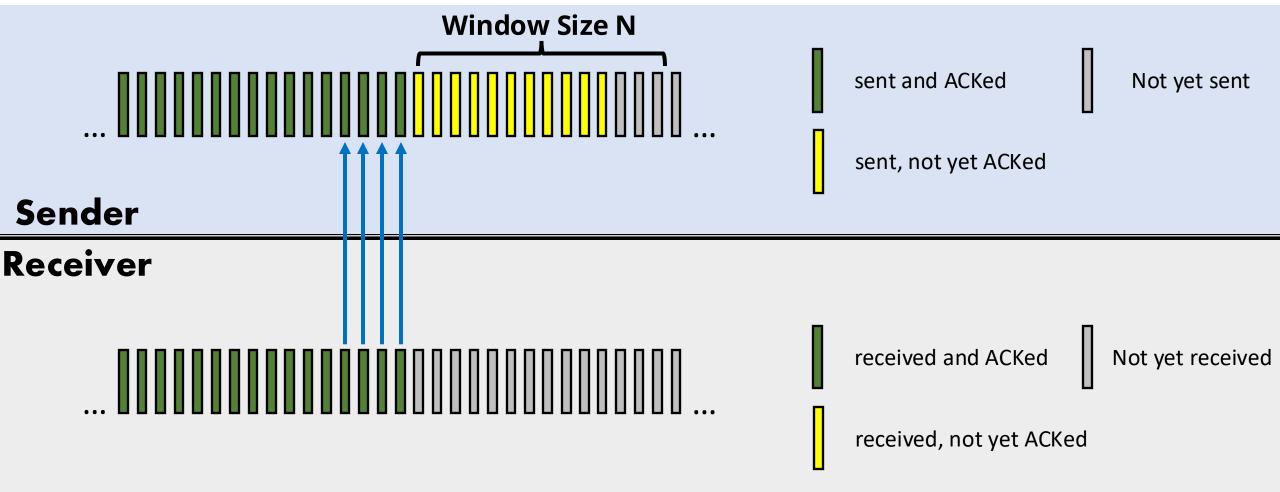
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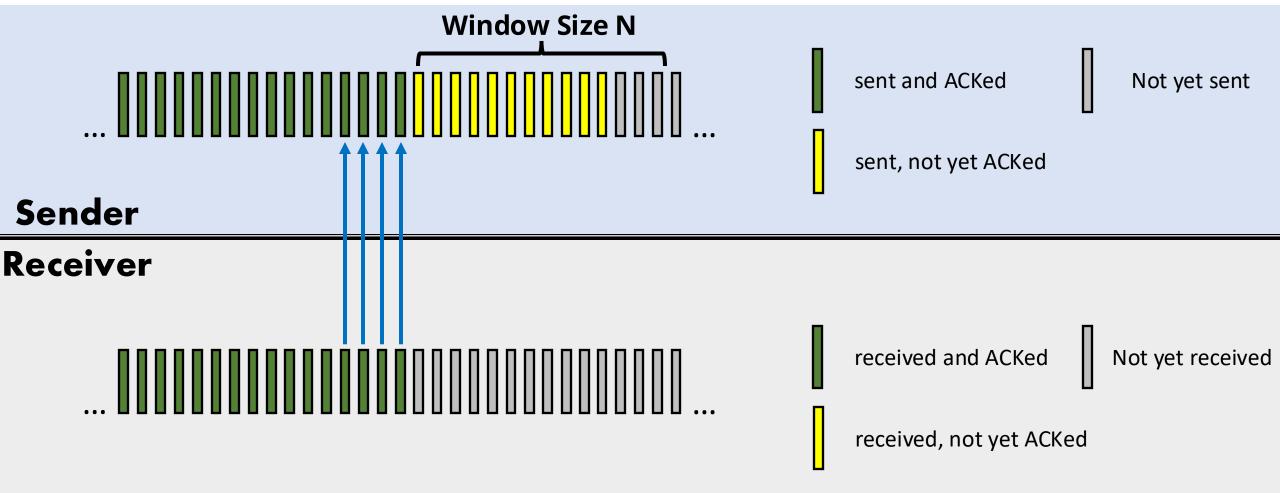
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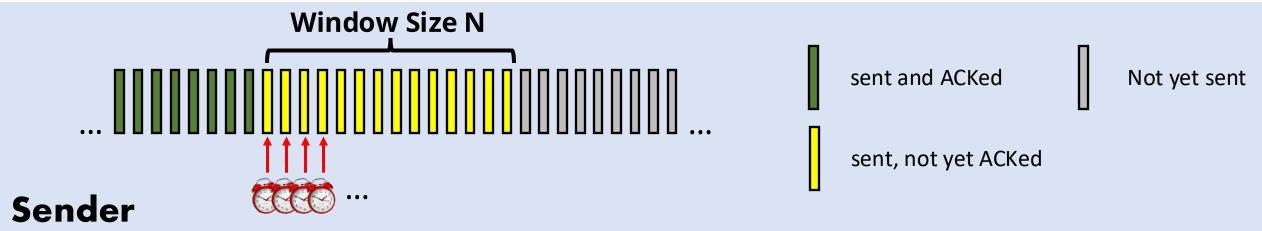
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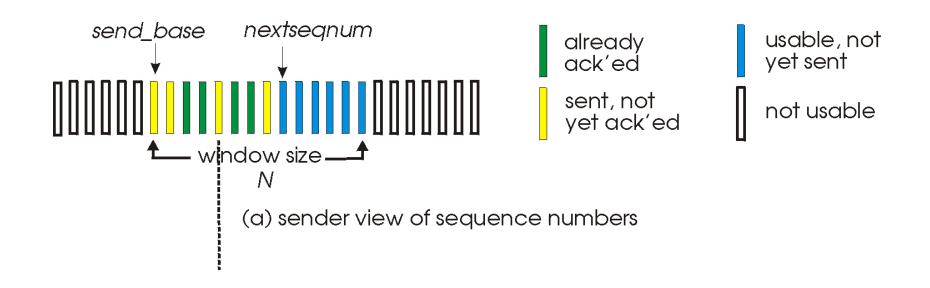


- Maintains (conceptually) a timer for each unACKed pkt
 - timeout: retransmits single unACKed packet associated with timeout

Selective repeat: the approach

- pipelining: multiple packets in flight
- receiver individually ACKs all correctly received packets
 - buffers packets, as needed, for in-order delivery to upper layer
- sender:
 - maintains (conceptually) a timer for each unACKed pkt
 - timeout: retransmits single unACKed packet associated with timeout
 - maintains (conceptually) "window" over N consecutive seq #s
 - limits pipelined, "in flight" packets to be within this window

Selective repeat: sender, receiver windows



Selective repeat: sender and receiver

sender

data from above:

• if next available seq # in window, send packet

timeout(*n*):

resend packet n, restart timer

ACK(n) in [sendbase,sendbase+N-1]:

- mark packet n as received
- if n smallest unACKed packet, advance window base to next unACKed seq #

receiver

packet n in [rcvbase, rcvbase+N-1]

- send ACK(n)
- out-of-order: buffer
- in-order: deliver (also deliver buffered, in-order packets), advance window to next not-yetreceived packet

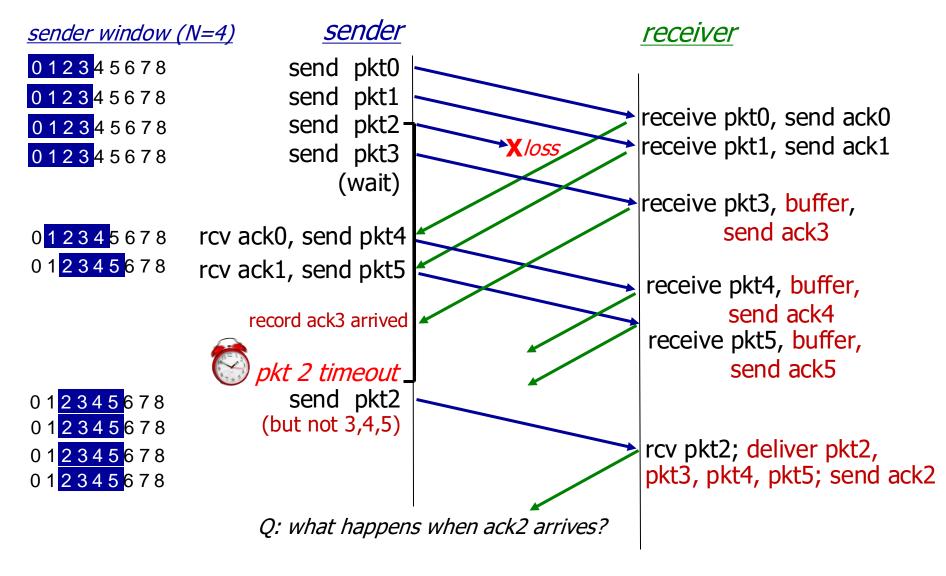
packet n in [rcvbase-N,rcvbase-1]

ACK(n)

otherwise:

ignore

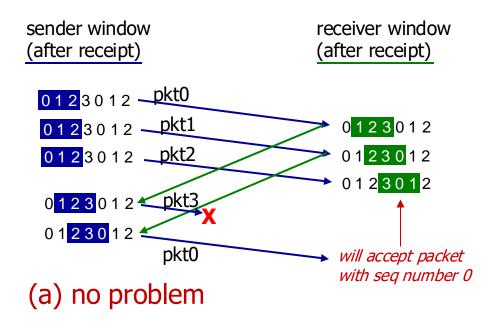
Selective Repeat in action

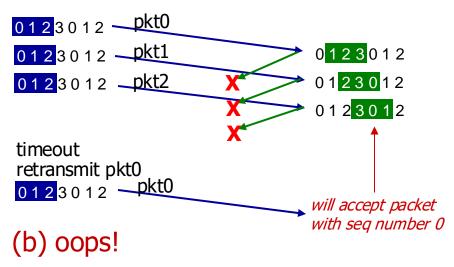


Selective repeat: a dilemma!

example:

- seq #s: 0, 1, 2, 3 (base 4 counting)
- window size=3

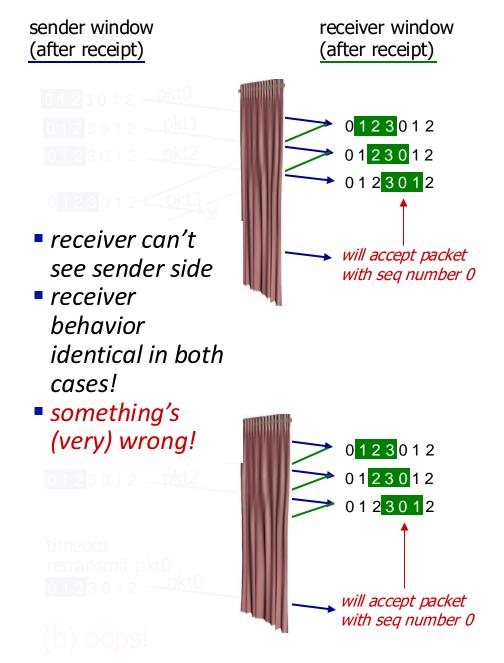




Selective repeat: a dilemma!

example:

- seq #s: 0, 1, 2, 3 (base 4 counting)
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Chapter 3: roadmap

- Transport-layer services
- Multiplexing and demultiplexing
- Connectionless transport: UDP
- Principles of reliable data transfer
- Connection-oriented transport: TCP
 - segment structure
 - reliable data transfer
 - flow control
 - connection management
- Principles of congestion control
- TCP congestion control



TCP: overview RFCs: 793,1122, 2018, 5681, 7323

- point-to-point:
 - one sender, one receiver
- reliable, in-order byte steam:
 - no "message boundaries"
- full duplex data:
 - bi-directional data flow in same connection
 - MSS: maximum segment size

- cumulative ACKs
- pipelining:
 - TCP congestion and flow control set window size
- connection-oriented:
 - handshaking (exchange of control messages) initializes sender, receiver state before data exchange
- flow controlled:
 - sender will not overwhelm receiver

TCP segment structure

32 bits source port # dest port # segment seq #: counting ACK: seq # of next expected bytes of data into bytestream sequence number byte; A bit: this is an ACK (not segments!) acknowledgement number length (of TCP header) receive window len used CE flow control: # bytes Internet checksum receiver willing to accept Urg data pointer checksum options (variable length) C, E: congestion notification TCP options application data sent by RST, SYN, FIN: connection data application into management (variable length) TCP socket

TCP sequence numbers, ACKs

Sequence numbers:

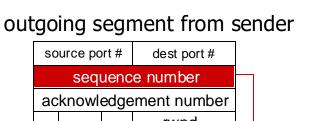
 byte stream "number" of first byte in segment's data

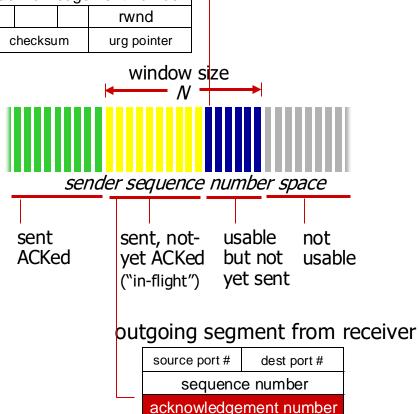
Acknowledgements:

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

Q: how receiver handles out-oforder segments

 <u>A:</u> TCP spec doesn't say, - up to implementor



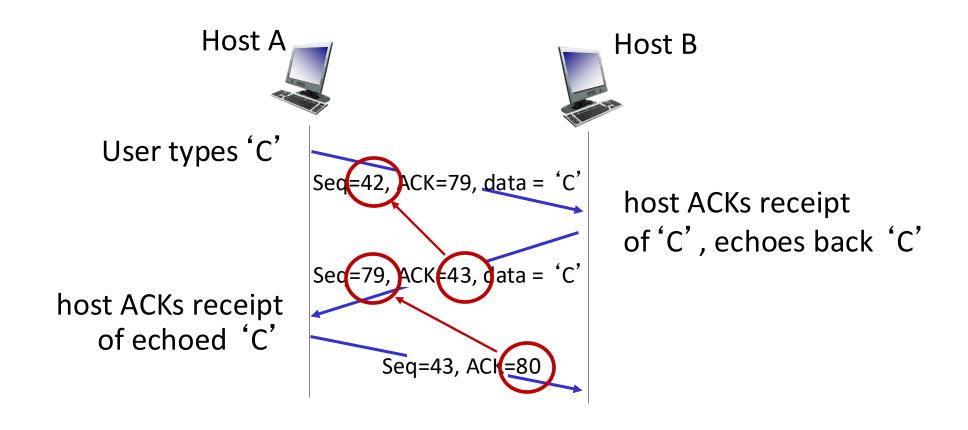


checksum

rwnd

urg pointer

TCP sequence numbers, ACKs



simple telnet scenario

TCP round trip time, timeout

- Q: how to set TCP timeout value?
- longer than RTT, but RTT varies!
- too short: premature timeout, unnecessary retransmissions
- too long: slow reaction to segment loss

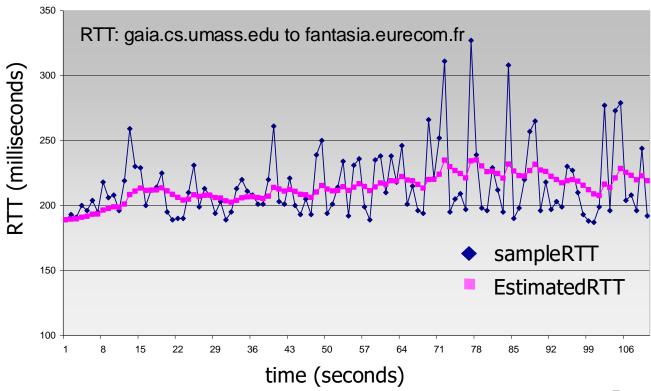
Q: how to estimate RTT?

- SampleRTT: measured time from segment transmission until ACK receipt
 - ignore retransmissions
- SampleRTT will vary, want estimated RTT "smoother"
 - average several recent measurements, not just current SampleRTT

TCP round trip time, timeout

EstimatedRTT = $(1-\alpha)$ *EstimatedRTT + α *SampleRTT

- <u>e</u>xponential <u>w</u>eighted <u>m</u>oving <u>a</u>verage (EWMA)
- influence of past sample decreases exponentially fast
- typical value: α = 0.125



TCP round trip time, timeout

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

■ DevRTT: EWMA of SampleRTT deviation from EstimatedRTT:

DevRTT =
$$(1-\beta)$$
*DevRTT + β *|SampleRTT-EstimatedRTT| (typically, $\beta = 0.25$)

^{*} Check out the online interactive exercises for more examples: http://gaia.cs.umass.edu/kurose_ross/interactive/

TCP Sender (simplified)

event: data received from application

- create segment with seq #
- seq # is byte-stream number of first data byte in segment
- start timer if not already running
 - think of timer as for oldest unACKed segment
 - expiration interval:TimeOutInterval

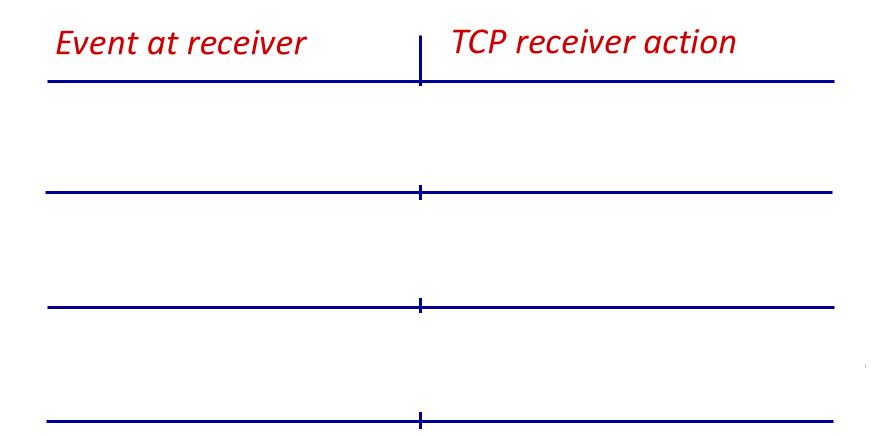
event: timeout

- retransmit segment that caused timeout
- restart timer

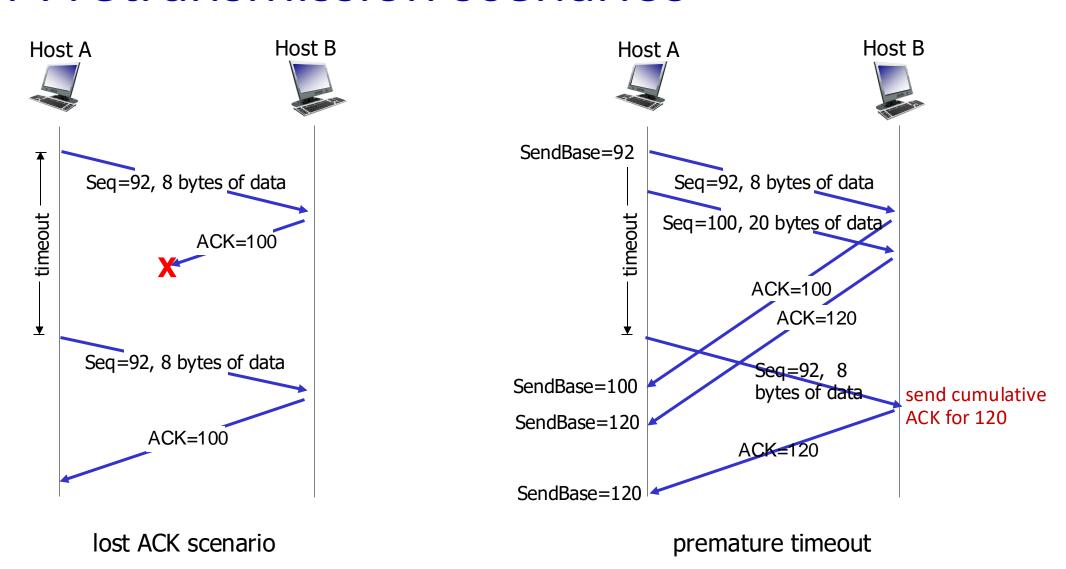
event: ACK received

- if ACK acknowledges previously unACKed segments
 - update what is known to be ACKed
 - start timer if there are still unACKed segments

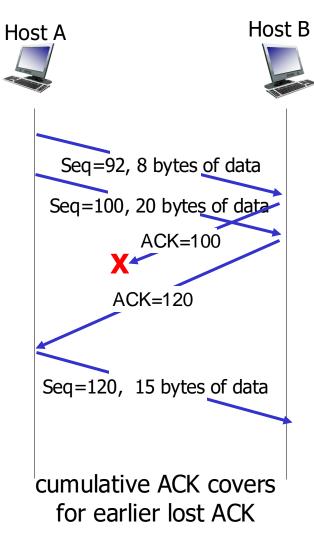
TCP Receiver: ACK generation [RFC 5681]



TCP: retransmission scenarios



TCP: retransmission scenarios



TCP fast retransmit

TCP fast retransmit

if sender receives 3 additional ACKs for same data ("triple duplicate ACKs"), resend unACKed segment with smallest seq #

 likely that unACKed segment lost, so don't wait for timeout

Receipt of three duplicate ACKs indicates 3 segments received after a missing segment – lost segment is likely. So retransmit!

