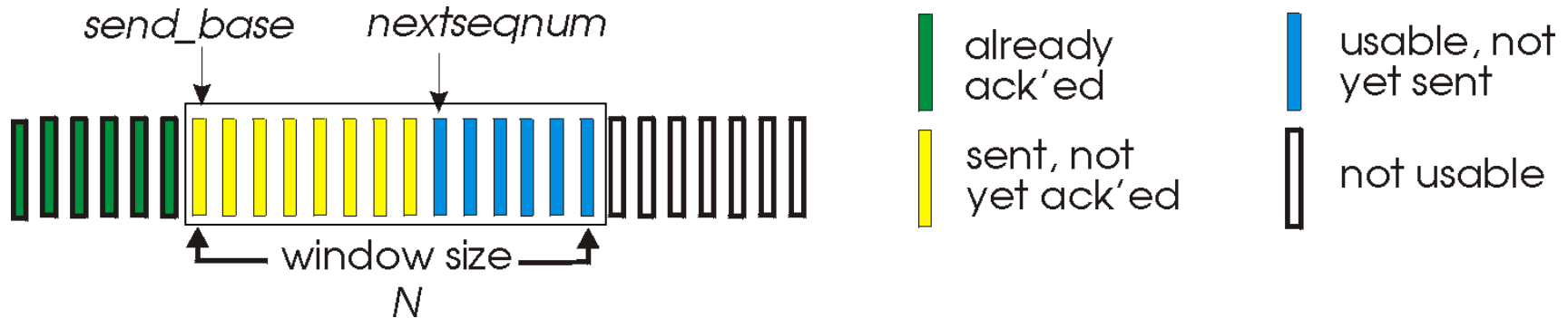


# Go-Back-N: sender

1. **Window**: up to N consecutive **unack'ed packets** allowed



2. **cumulative ACK**(n): all packets with a sequence number up to and including n have been correctly received at the receiver
  - may receive duplicate ACKs (see receiver)
3. **timer** for **oldest in-flight pkt** (i.e., **send\_base**)
  - ❖ **timeout**: retransmit packets from send\_base to nextseqnum-1 (i.e., yellow region)

# GBN in action

sender window (N=4)

0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8  
 0 1 2 3 4 5 6 7 8

sender

send pkt0  
 send pkt1  
 send pkt2  
 send pkt3  
 (wait)

rcv ack0, send pkt4  
 rcv ack1, send pkt5

ignore duplicate ACK



*pkt 2 timeout*

send pkt2  
 send pkt3  
 send pkt4  
 send pkt5

receiver

receive pkt0, send ack0  
 receive pkt1, send ack1

receive pkt3, discard,  
 (re)send ack1

receive pkt4, discard,  
 (re)send ack1

receive pkt5, discard,  
 (re)send ack1

rcv pkt2, deliver, send ack2  
 rcv pkt3, deliver, send ack3  
 rcv pkt4, deliver, send ack4  
 rcv pkt5, deliver, send ack5

# Selective repeat

- ❖ receiver *individually* acknowledges all correctly received pkts
  - buffers pkts, as needed, for eventual in-order delivery to upper layer
- ❖ sender only resends pkts for which ACK not received
  - sender timer for each unACKed pkt
- ❖ sender window
  - $N$  consecutive seq #'s
  - limits seq #s of sent, unACKed pkts

# Selective repeat

## sender

### data from above:

- ❖ if next available seq # in window, send pkt

### timeout(n):

- ❖ resend pkt n, restart timer

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

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

## receiver

### pkt n in [rcvbase, rcvbase+N-1]

- ❖ send ACK(n)
- ❖ out-of-order: buffer
- ❖ in-order: deliver (also deliver buffered, in-order pkts), advance window to next not-yet-received pkt

### pkt n in [rcvbase-N, rcvbase-1]

- ❖ ACK(n)

### otherwise:

- ❖ ignore

# Selective repeat in action

sender window (N=4)

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

0 1 2 3 4 5 6 7 8

sender

send pkt0

send pkt1

send pkt2

send pkt3

(wait)

rcv ack0, send pkt4

rcv ack1, send pkt5

record ack3 arrived



*pkt 2 timeout*

send pkt2

record ack4 arrived

record ack5 arrived

receiver

receive pkt0, send ack0

receive pkt1, send ack1

receive pkt3, buffer,  
send ack3

receive pkt4, buffer,  
send ack4

receive pkt5, buffer,  
send ack5

rcv pkt2; deliver pkt2,  
pkt3, pkt4, pkt5; send ack2

*Q: what happens when ack2 arrives?*

# Problem I

a) For GBN, the ACK is cumulative. Thus, the ACKs that the sender will receive are

ACK0, ACK1, ACK2, ACK3, ACK4, ACK5, ACK5.

Note: For the lost pkt6, the receiver will not send any ACK back. The last ACK5 is the response of receiving pkt7.

b) For SR, only correctly received packets will be individually acknowledged, while all other cases will be ignored. Thus, the ACKs that the sender will receive are

ACK0, ACK1, ACK2, ACK3, ACK4, ACK5, ACK7.

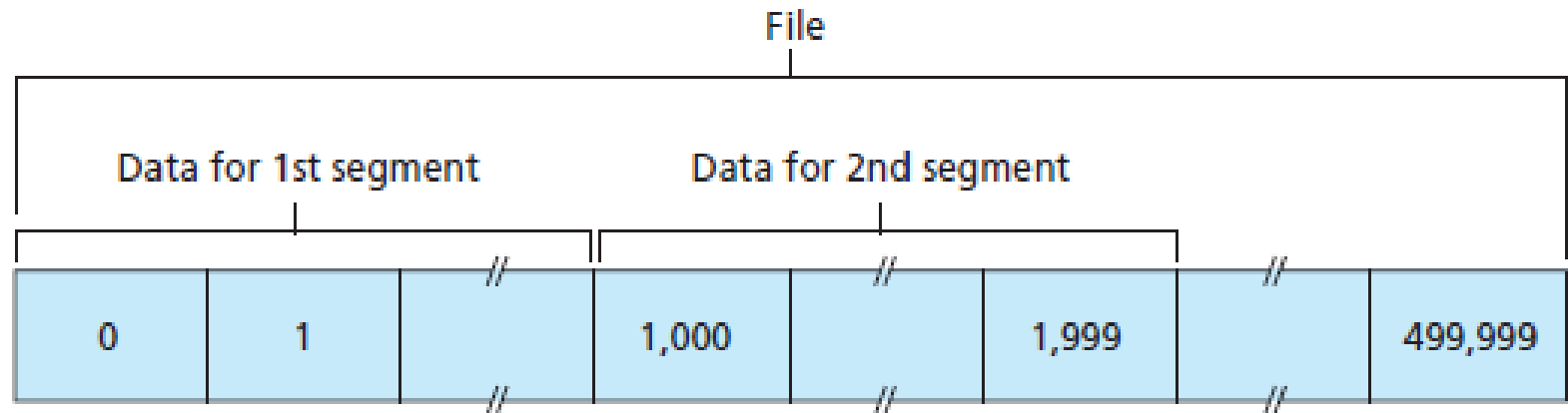
Note: since pkt7 is out-of-order, the receiver will buffer it until pkt6 is eventually correctly received (from retransmission by sender).

# TCP payload

- ❖ A TCP header (typically 20-60 bytes long), which includes metadata like source/destination ports, sequence numbers (to track the order of bytes in the data stream), acknowledgment numbers (for confirming receipt), flags (e.g., SYN, ACK, FIN), and checksums for error detection.
- ❖ A payload (the actual data being transferred), which is limited in size by the Maximum Segment Size (MSS). The MSS is the maximum amount of data that can fit into one segment (excluding the header), often negotiated during connection setup to avoid fragmentation. For example, a common MSS is 1,460 bytes on Ethernet networks.

# TCP seq. numbers

- ❖ **Maximum segment size (MSS)**: A TCP **payload** (the actual data being transferred), which is limited in size by the Maximum Segment Size (MSS).
- ❖ **Sequence number** for a segment: Byte-stream number of the first byte in the segment
- ❖ E.g., file size = 500,000 bytes and MSS = 1,000 bytes



Dividing file data into TCP segments



# TCP ACKs

- ❖ ACK number that receiver puts in its segment  
= seq # of next byte expected from other side
  - E.g., Host A has received all bytes 0 through 535 from B  
-> ACK number = 536
  - E.g., Host A has received one segment from Host B containing bytes 0 through 535 and another segment containing bytes 900 through 1,000  
-> ACK number = 536  
(acknowledges bytes up to the first missing byte of 536)
- ❖ Cumulative ACK (GBN-style by default): TCP acknowledges bytes *up to the first missing byte* in the stream
- ❖ Q: how receiver handles *out-of-order* segments
  - A: TCP spec doesn't say, - up to implementor

## Problem 2

- TCP treats data as a continuous byte stream, where each segment's sequence number (seq) indicates the **position of its first byte** in that stream.
- The acknowledgment number (ack) in a response segment tells the sender "I've **received everything up to (but not including)** this byte number."
- Host B acknowledges every incoming segment, but the **ack number depends on whether the data is in-order.**

## Problem 2

- Given Setup
  - Host B has received and acknowledged all bytes 0–126. So, its current "next expected byte" is 127.
  - Host A sends Segment 1: 80 bytes of data, seq = 127 (starts with byte 127, covers bytes 127–206).
  - Host A sends Segment 2 (back-to-back): 40 bytes of data (covers bytes 207–246).
  - Ports in Segment 1: source (A) = 302, destination (B) = 80.

## Problem 2.a

Sequence number, source port, and destination port in the second segment (from A to B)

- a. Sequence number: This is always the byte position of the first byte in the segment. Segment 1 covers bytes 127–206 (80 bytes), so Segment 2 starts right after at byte 207. Thus, seq = 207.
- b. Source port: This doesn't change—it's still Host A's port, 302.
- c. Destination port: Still Host B's port, 80.

Correct answer: D) 207, 302, 80

## Problem 2.b

If the first segment arrives before the second, what is the ack number, source port, and destination port in the acknowledgment of the first arriving segment (from B to A)?

- Acknowledgment number: Host B receives Segment 1 (bytes 127–206) in order (since it was expecting 127). It now has everything up to byte 206, so it acknowledges the next expected byte: 207.
- Source port: In the ACK (from B to A), ports swap—B's original destination port (80) becomes its source port: 80. Destination port: B sends back to A's original source port: 302.

**Correct answer: C) 207, 80, 302**

## Problem 2.c

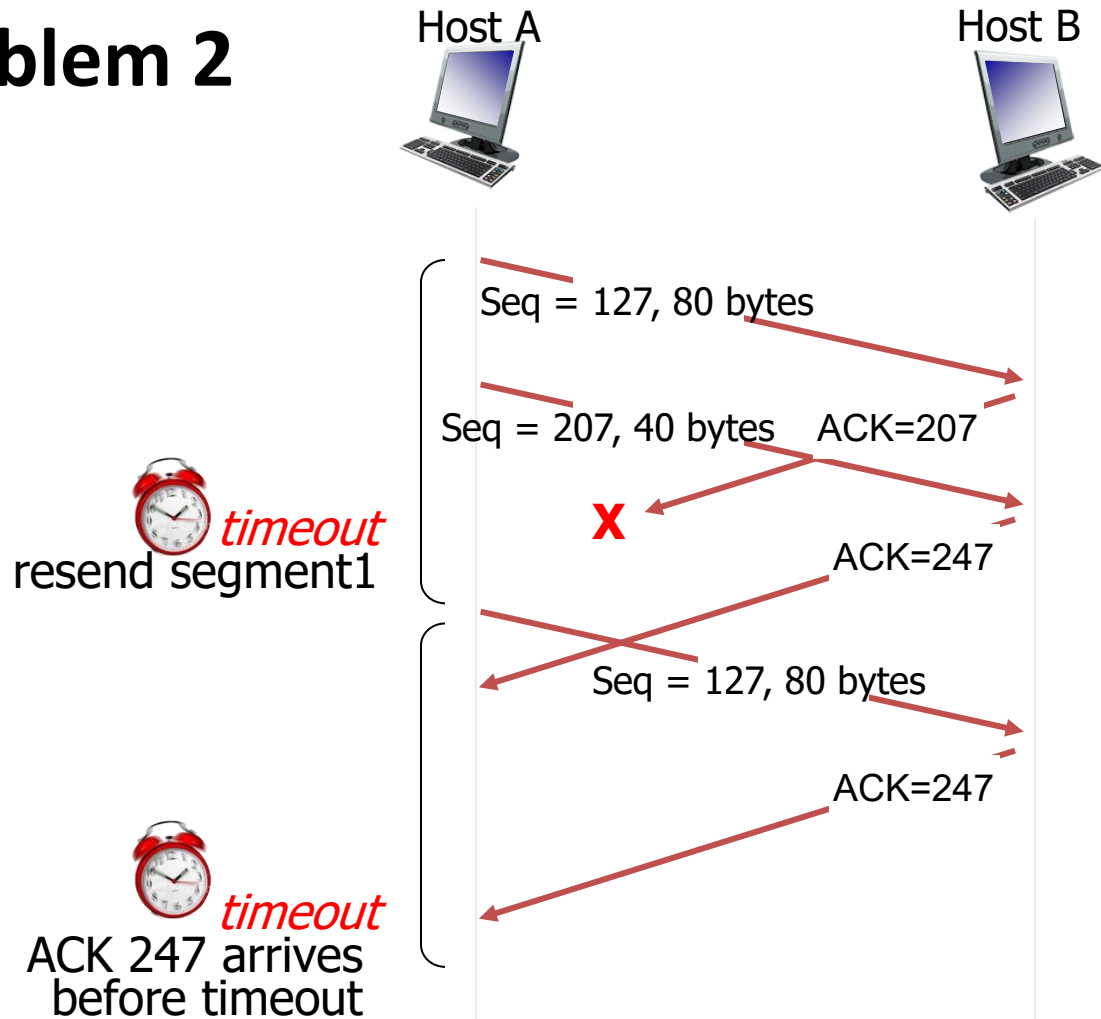
If the second segment arrives before the first, what is the acknowledgment number in the acknowledgment of the first arriving segment (from B to A)?

- Acknowledgment number: Host B receives Segment 2 first (bytes 207–246), but it's out of order—B is still expecting byte 127. TCP buffers out-of-order segments but doesn't advance its ack number until the gap is filled. So, B acknowledges its current next expected byte, which remains 127. (This prompts A to retransmit Segment 1 if needed.)

Correct answer: A) 127      This demonstrates TCP's reliable, ordered delivery: **acks are cumulative**

# Problem 2

d)



## Timeout Interval for re-transmission

- In TCP, the retransmission timeout (RTO) interval—often called the TimeoutInterval—is adaptively computed to balance **responsiveness** and **stability** in the face of varying network conditions.
  - Given the values of EstimatedRTT (the smoothed estimate of the round-trip time) and DevRTT (the mean deviation of the RTT, capturing variability)



## Problem 3

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



↑  
estimated RTT

↑  
“safety margin”

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

(typically,  $\alpha = 0.125$ )

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

(typically,  $\beta = 0.25$ )

The multiplier of 4 is empirically derived to set the timeout conservatively—about 4 standard deviations above the mean RTT

## Problem 4

After obtaining the first sampleRTT 106 ms:

- EstimatedRTT =  $(1 - 0.125) \cdot 100 + 0.125 \cdot 106 = 0.875 \cdot 100 + 13.25 = 100.75$  ms
- DevRTT =  $(1 - 0.25) \cdot 5 + 0.25 \cdot |106 - 100.75| = 0.75 \cdot 5 + 0.25 \cdot 5.25 = 3.75 + 1.3125 = 5.0625$  ms
- TimeoutInterval =  $100.75 + 4 \cdot 5.0625 = 121$  ms

## Problem 4

After obtaining the second sample RTT 120 ms:

- EstimatedRTT =  $0.875 \cdot 100.75 + 0.125 \cdot 120 = 88.15625 + 15 = 103.15625$  ms
- DevRTT =  $0.75 \cdot 5.0625 + 0.25 \cdot |120 - 103.15625| = 3.796875 + 4.2109375 = 8.0078125$  ms
- TimeoutInterval =  $103.15625 + 4 \cdot 8.0078125 = 103.15625 + 32.03125 = 135.1875$  ms

## Problem 4

After obtaining the third sampleRTT 140ms:

- EstimatedRTT =  $0.875 \cdot 103.15625 + 0.125 \cdot 140 = 90.26171875 + 17.5 = 107.76171875$  ms
- DevRTT =  $0.75 \cdot 8.0078125 + 0.25 \cdot |140 - 107.76171875| = 6.005859375 + 8.0595703125 = 14.0654296875$  ms
- TimeoutInterval =  $107.76171875 + 4 \cdot 14.0654296875 = 107.76171875 + 56.26171875 = 164.0234375$  ms