

Transport Layer Part 7

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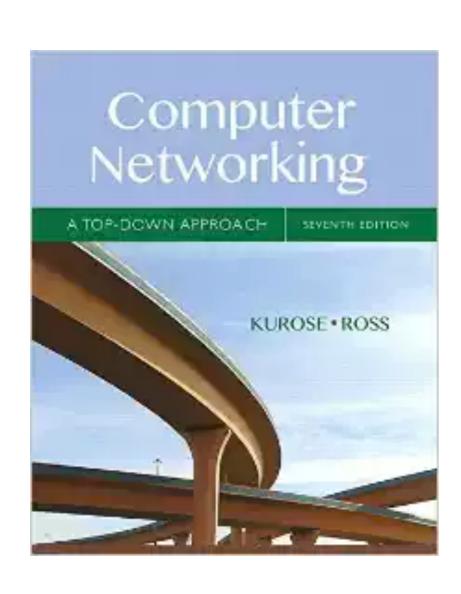
"Wednesday just don't go ... Thursday goes too slow ..."

These slides are more-or-less directly from the slide set developed by Jim Kurose and Keith Ross for their book "Computer Networking: A Top Down Approach, 5th edition".

The slides have been lightly adapted for Mark Allman's EECS 325/425 Computer Networks class at Case Western Reserve University.

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Reading Along ...



- 3.5: Connectionoriented transport: TCP
 - reliable data transfer

- *time-out period
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 - sender often sends many segments back-toback
 - if segment is lost, there will likely be many duplicate ACKs.

- * if sender receives 3
 ACKs for the same data,
 it supposes that segment
 after ACKed data was
 lost:
 - fast retransmit: resend segment before timer expires

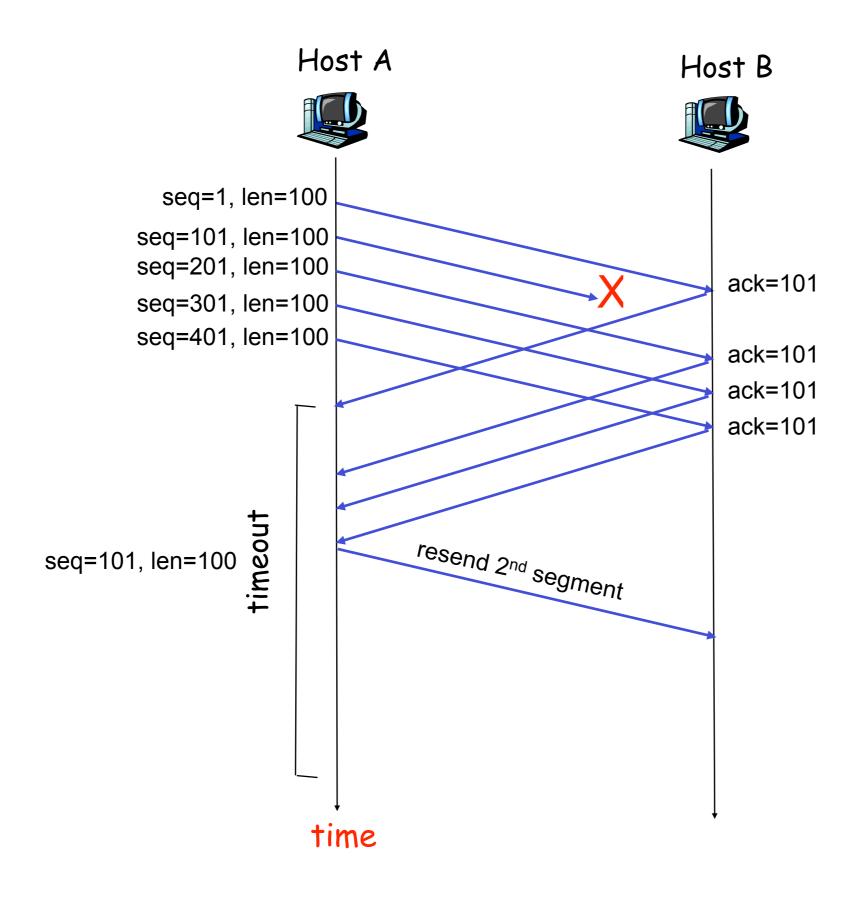


Figure 3.37 Resending a segment after triple duplicate ACK

Fast retransmit algorithm:

```
event: ACK received, with ACK field value of y
          if (y > SendBase) {
             SendBase = y
             if (there are currently not-yet-acknowledged segments)
                 start timer
          else {
               increment count of dup ACKs received for y
               if (count of dup ACKs received for y = 3) {
                  resend segment with sequence number y
```

a duplicate ACK for already ACKed segment

fast retransmit

SR or GBN?

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*Is this selective repeat?

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- *To get something akin to selective repeat we use the Selective Acknowledgment (SACK) option
 - Indicates blocks of (begseq,endseq) sequence numbers that have arrived

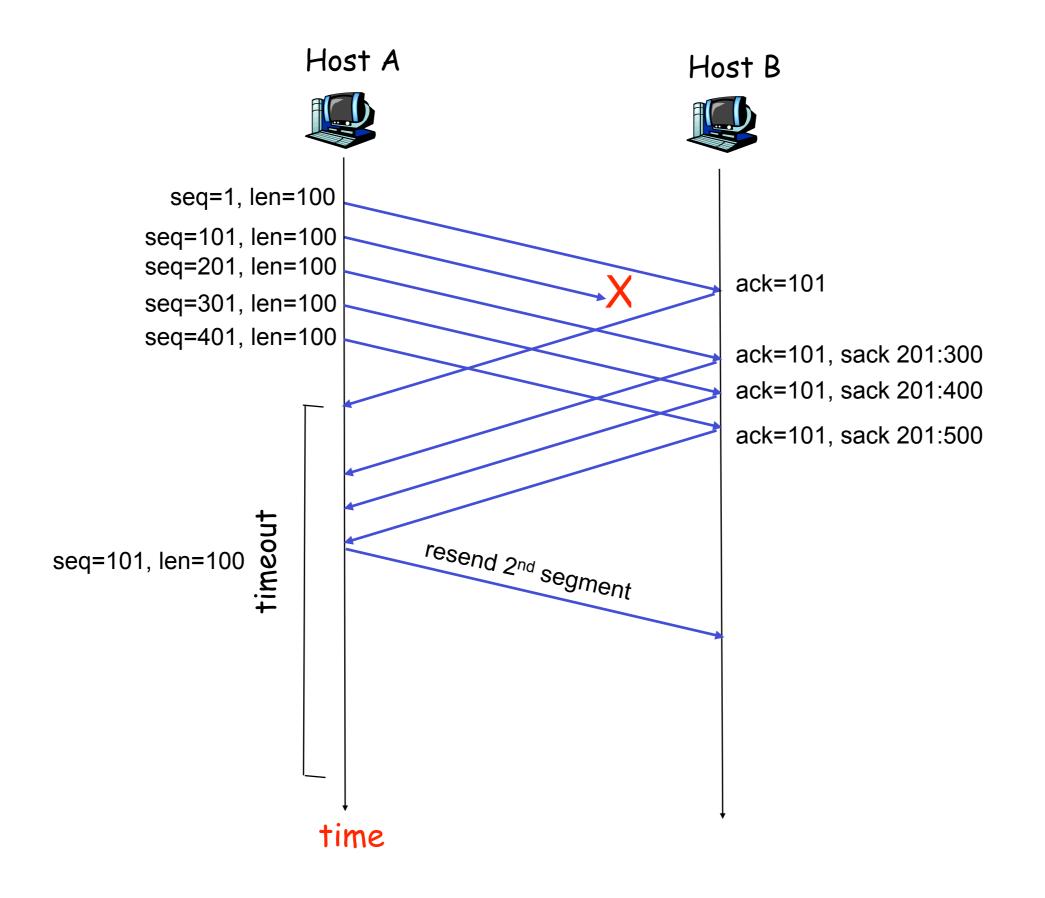
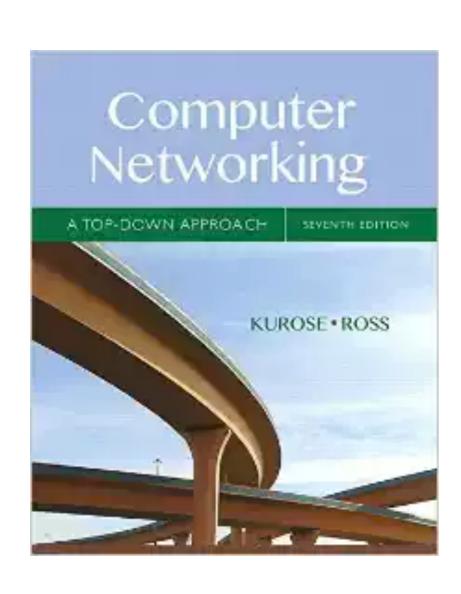


Figure 3.37 Resending a segment after triple duplicate ACK

SACK

- Fundamental change
 - cumulative ACK allows us to focus on only a single packet at a time
 - SACK allows us to concentrate on multiple packets at a time
 - · i.e., the reality of a pipelined system
- *SACK algorithms get complicated
 - see RFC 6675 for an example

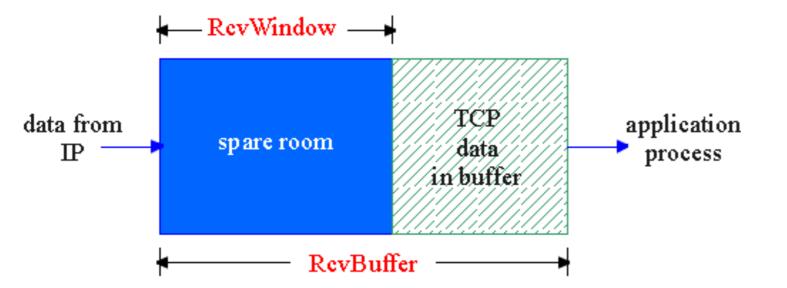
Reading Along ...



- 3.5: Connectionoriented transport: TCP
 - flow control

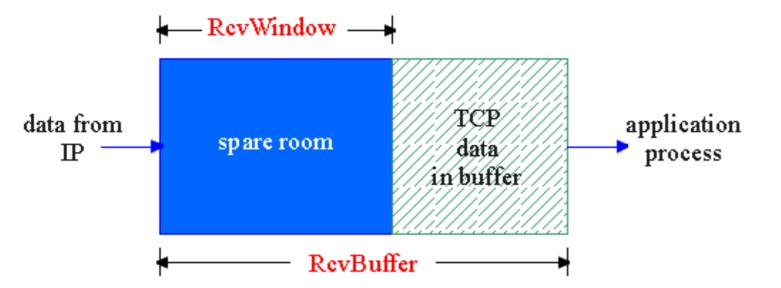
TCP Flow Control

*receive side of TCP connection has a receive buffer:



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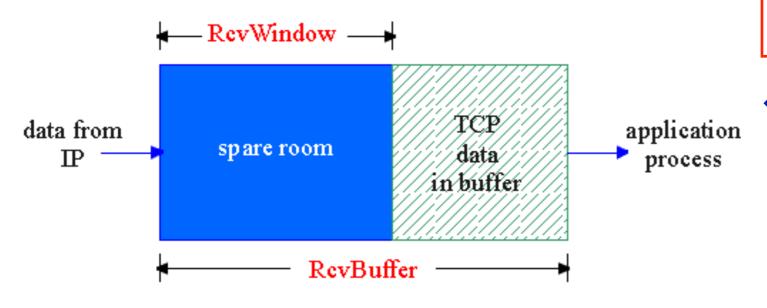
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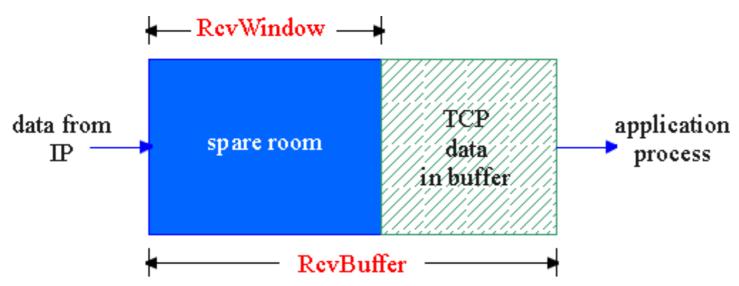
app process may be slow at reading from buffer

flow control

sender won't overflow receiver's buffer by transmitting too much, too fast

* speed-matching service: matching the send rate to the receiving app's drain rate

TCP Flow control: how it works



(suppose TCP receiver discards out-of-order segments)

- *spare room in buffer
- = RcvWindow

- *rcvr advertises spare room by including value of RcvWindow in segments
- *sender limits unACKed data to RcvWindow
 - guarantees receive buffer doesn't overflow

bit _	1 2 3	4 5 6 7 8 5	10	11	12	13	14	15	16	17	18	19	20	21	22	23	24	25	26	27	28	29	30	31
	Source Port													Des	tinat	ion	Port							
	Sequence Number																							
					Д	kckn	owle	dge	mei	nt Nu	mbe	r												
	HLEN	Reserved	U R G	A C K	P 8 H	R S T	S Y N	FIN	Window															
	Checksum								Urgent Pointer															
	Options (if any)					Padding																		
	Data																							
T.I.																								

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- *Is 64KB enough?
 - No!
 - This is a place where TCP's original design has been eclipsed by reality

*ReqAdvWindow = BW x RTT

*Or, BW = Window / RTT

We need to be able to fill the so-called "bandwidth-delay product" of a network path to fully utilize the capacity of the path

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Many common scenarios—and some uncommon ones!—require windows TCP cannot encode.

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- *Must be agreed to in the SYNACK

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*Receiver:

- received_win = 10001100 00000000 = 35840
- true_window = received_win << 2
 = 10 00110000 00000000 = 140 KB</pre>

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- *Accommodates most (all?) current network paths
 - some are currently pushing to support use of 15 bits of scaling and hence windows with 2^31 or 2GB