

Error and Flow Control in TCP

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TCP Reliable Data Transfer



- Network layer service is unreliable, i.e. **does not guarantee**
 - datagram delivery
 - in-order delivery
 - data integrity
- TCP's reliable data transfer (RDT) service **ensures** that
 - the data stream that a process reads out of its TCP receive buffer is **uncorrupted, without gaps, without duplication, and in sequence**

TCP Error Control



- Error control in TCP is done by :
 - checksum
 - ACK
 - time-out
- TCP sends an ACK for the next packet that it is expecting
- When does a receiver generate ACK?
 - Rule-1: when node A sends data to node B, it piggybacks ACK
 - Rule-2: the receiver has no data to send and it receives an in-order segment, it delays sending ACK
 - Rule-3: there should not be more than two in-order unACKed segments at any time (it is delayed ACK)
 - Rule-4: when a segment arrives with an out-of-order sequence number; or, it is fast retransmission of missing segments
 - Rule-5: when a missing segment arrives
 - Rule-6: If a duplicate segment arrives

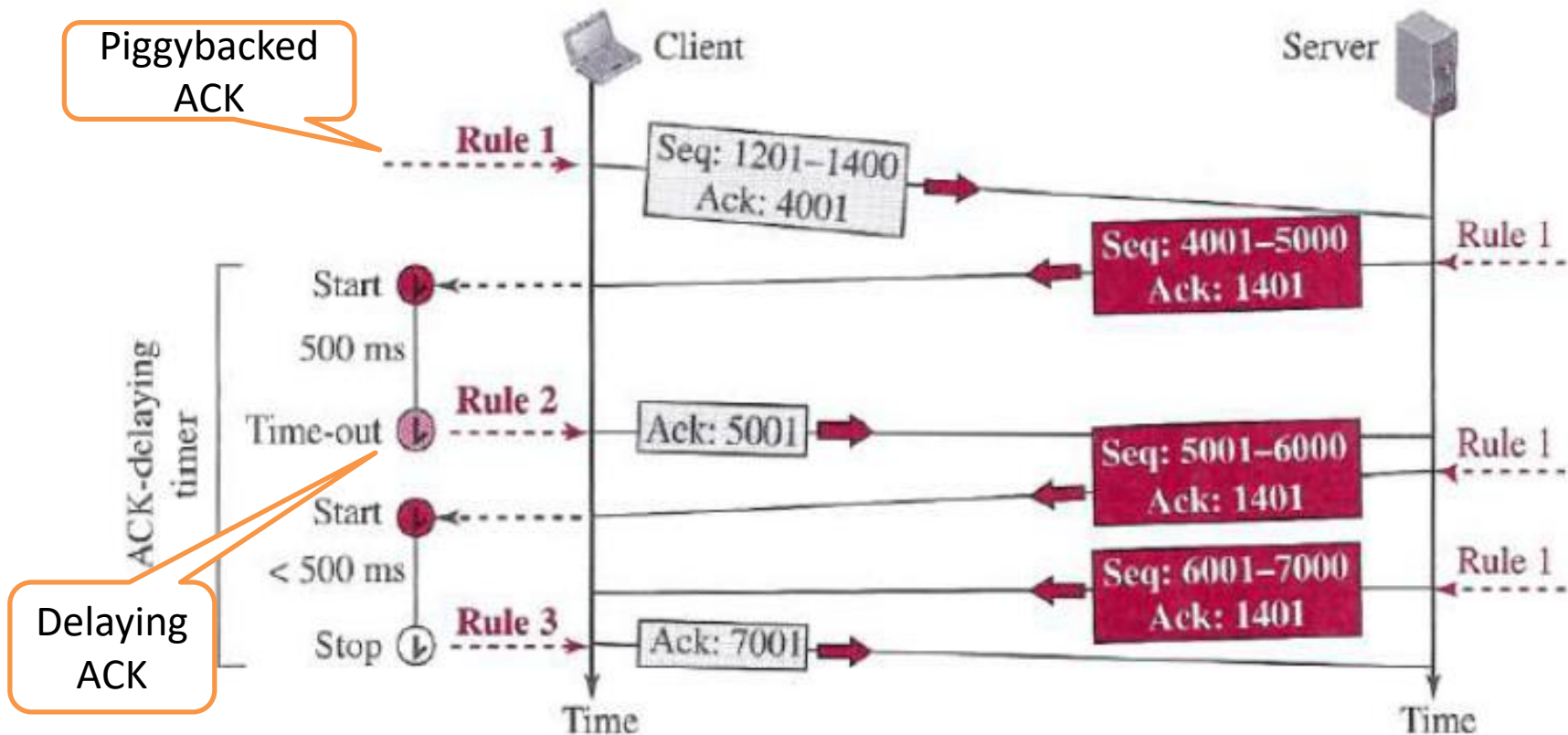
Cont...



- When retransmission happens?
 - After time-out
 - sending TCP maintains one retransmission time-out (RTO) for each connection
 - Three duplicate ACK rule
 - if three duplicate ACK (i.e., an original ACK and three exactly identical copies) arrive for a segment, the next segment is retransmitted without waiting for the time-out.
- Why three duplicate ACK?
 - Since TCP does not know whether a duplicate ACK is caused by a lost segment or just a reordering of segments, it waits for at max two duplicate ACK.
 - If three or more duplicate ACKs are received in a row, it is a strong indication that a segment has been lost.
- Data may arrive out-of-order and be temporarily stored by the receiving TCP.
- But, no out-of-order data are delivered to the process by TCP.

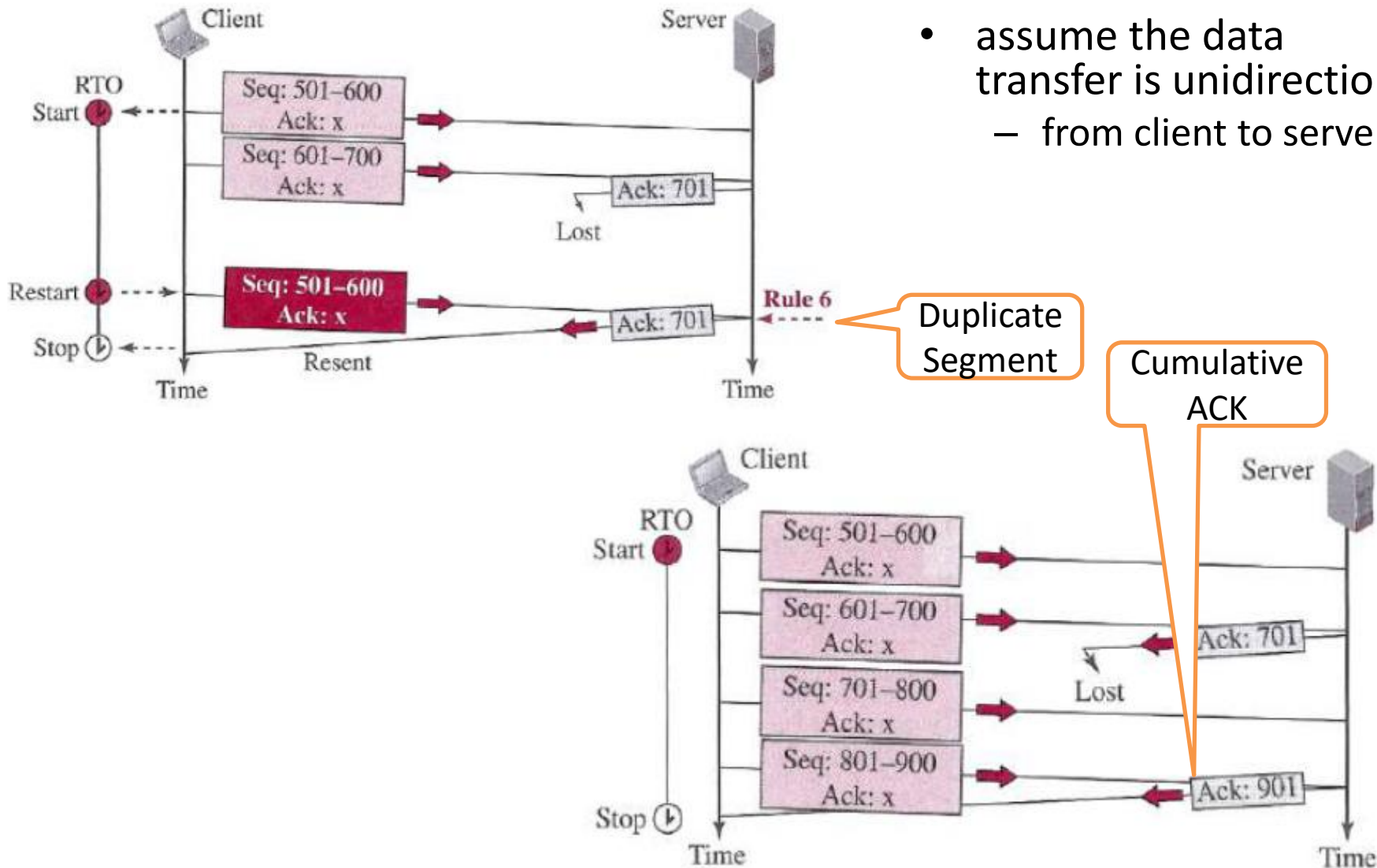
Example : Normal Scenario

- client TCP sends **one** segment (2000 byte);
- server TCP sends **three** segments (3 x 1000 byte).

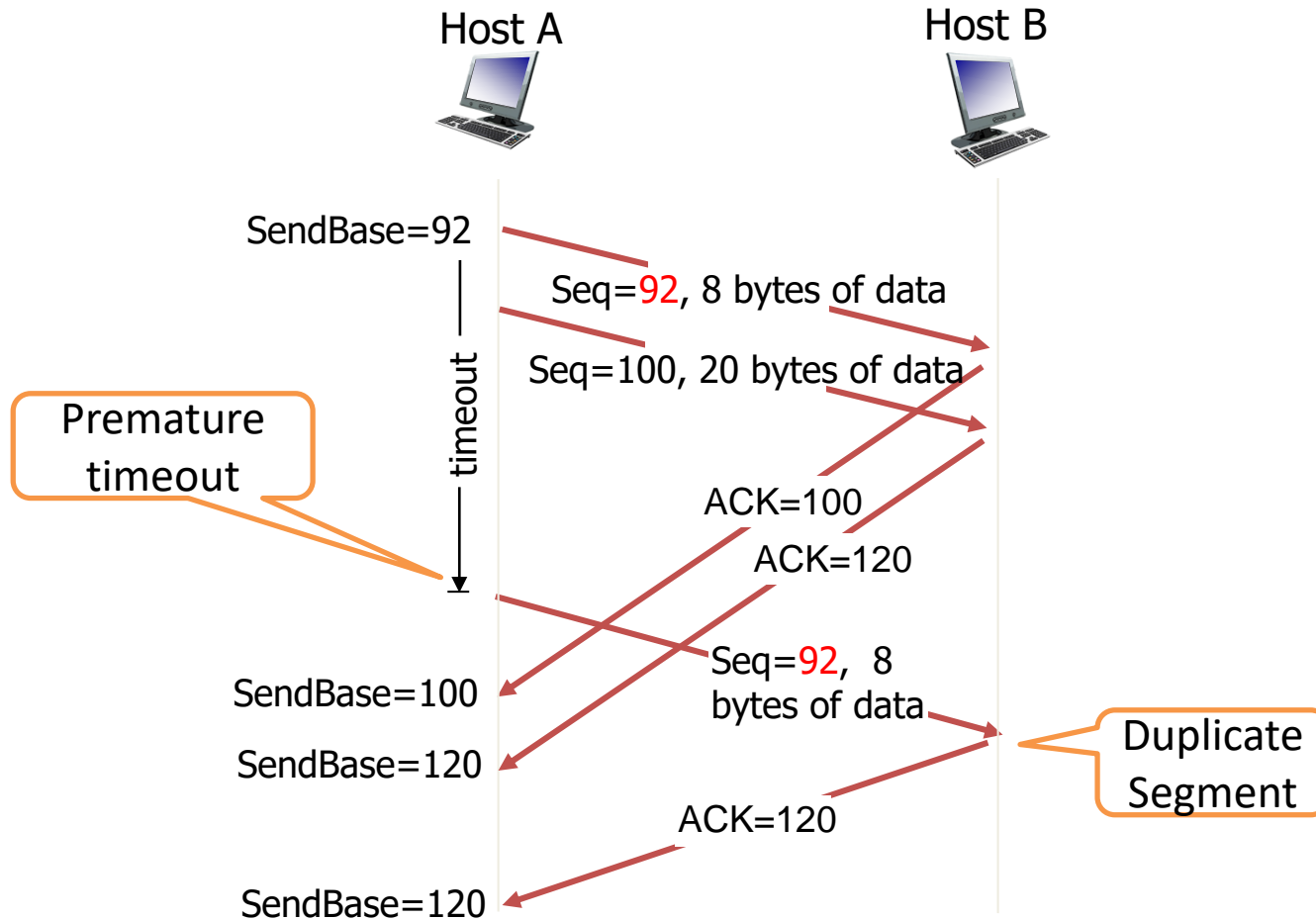


Example : Lost ACK, Cumulative ACK

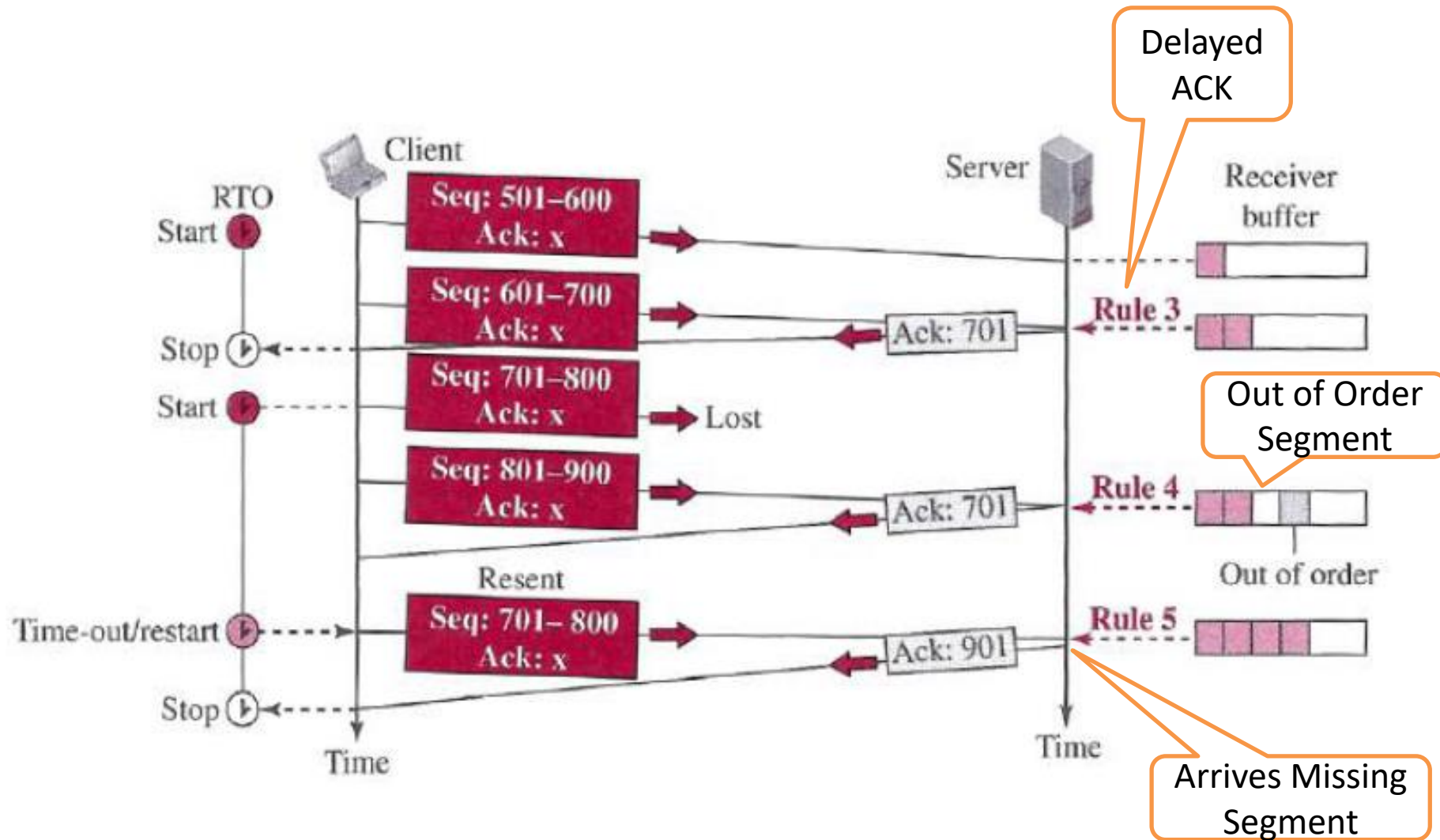
- assume the data transfer is unidirectional – from client to server



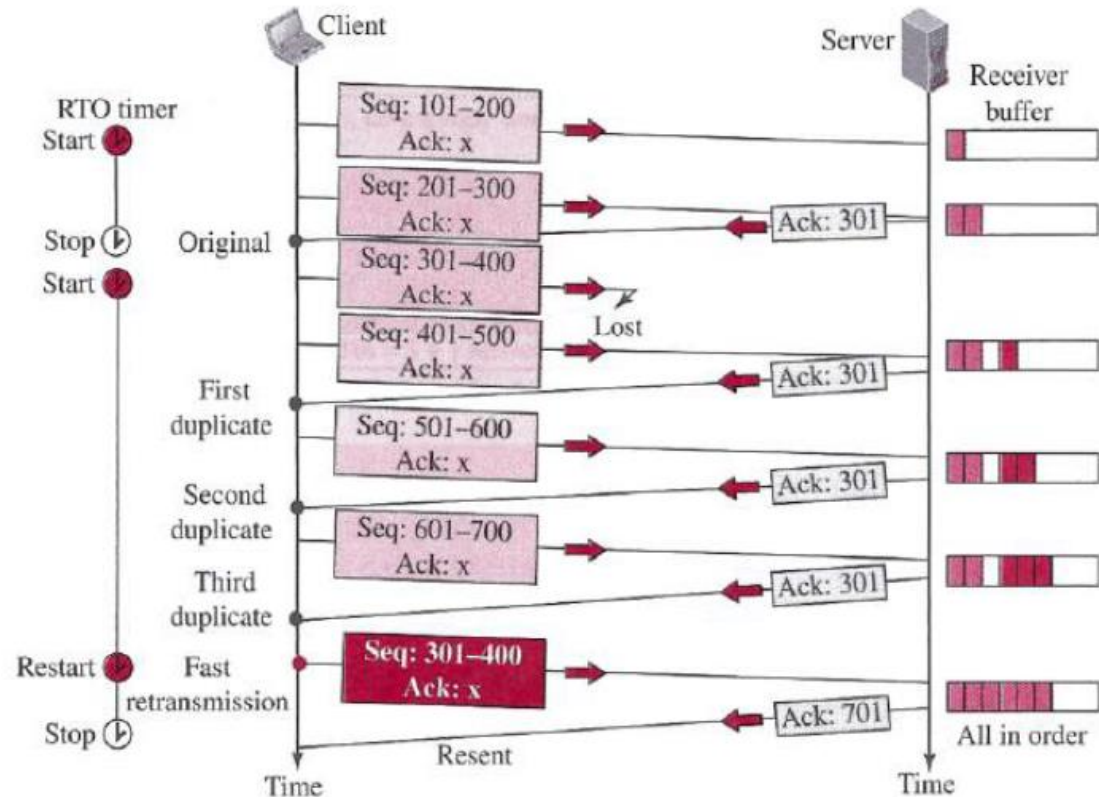
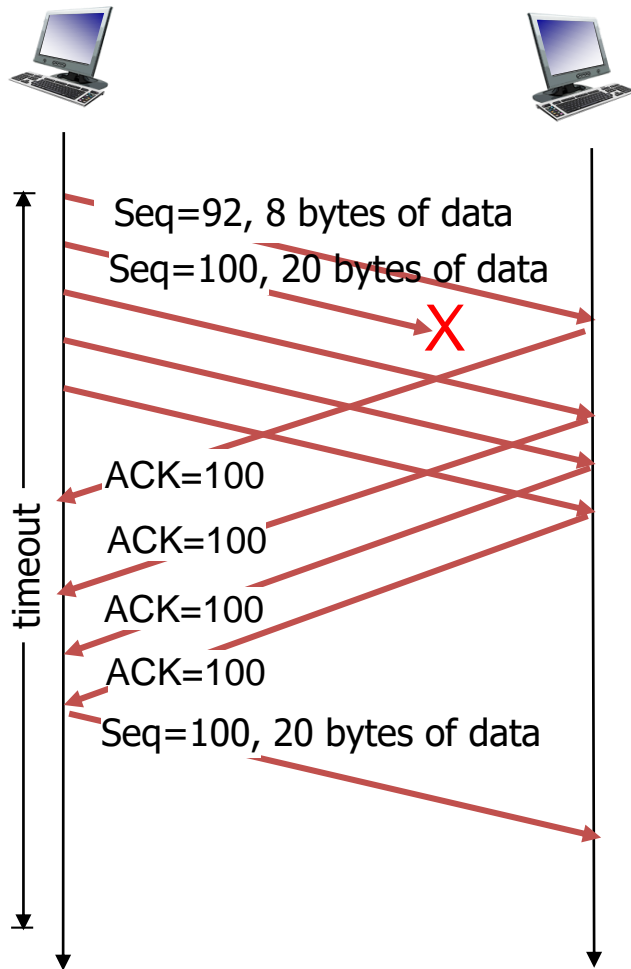
Example : Premature timeout



Example : Lost Segment



Example : Fast Retransmission after 3 duplicates

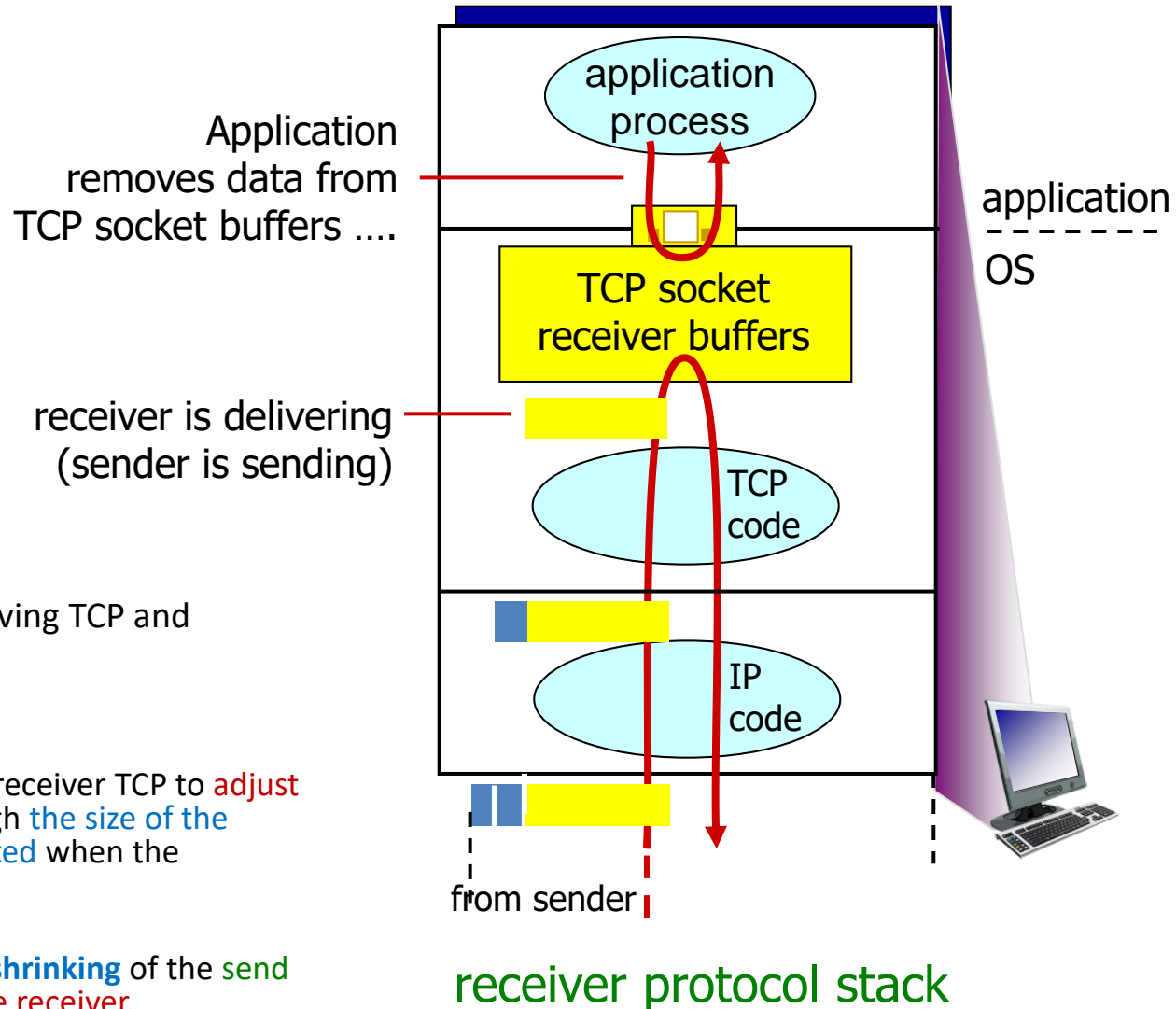


TCP Flow Control

flow control

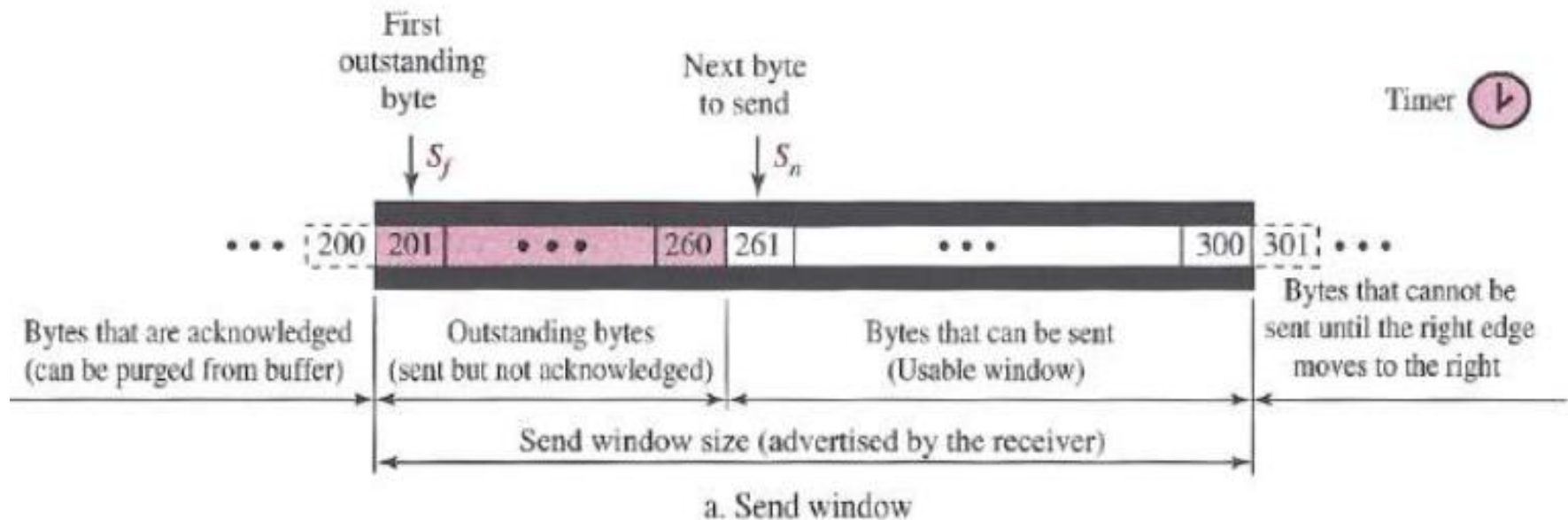
receiver controls
sender, so **sender won't
overflow receiver's
buffer by transmitting
too much, too fast**

- No flow control between receiving TCP and receiving process.
- To **achieve flow control**:
 - forces the sender TCP and receiver TCP to **adjust their window sizes**, although **the size of the buffer for both parties is fixed** when the connection is established.
 - The **opening, closing, and shrinking** of the **send window** is **controlled by the receiver**.



Send Window in TCP

- Let send window size = 100

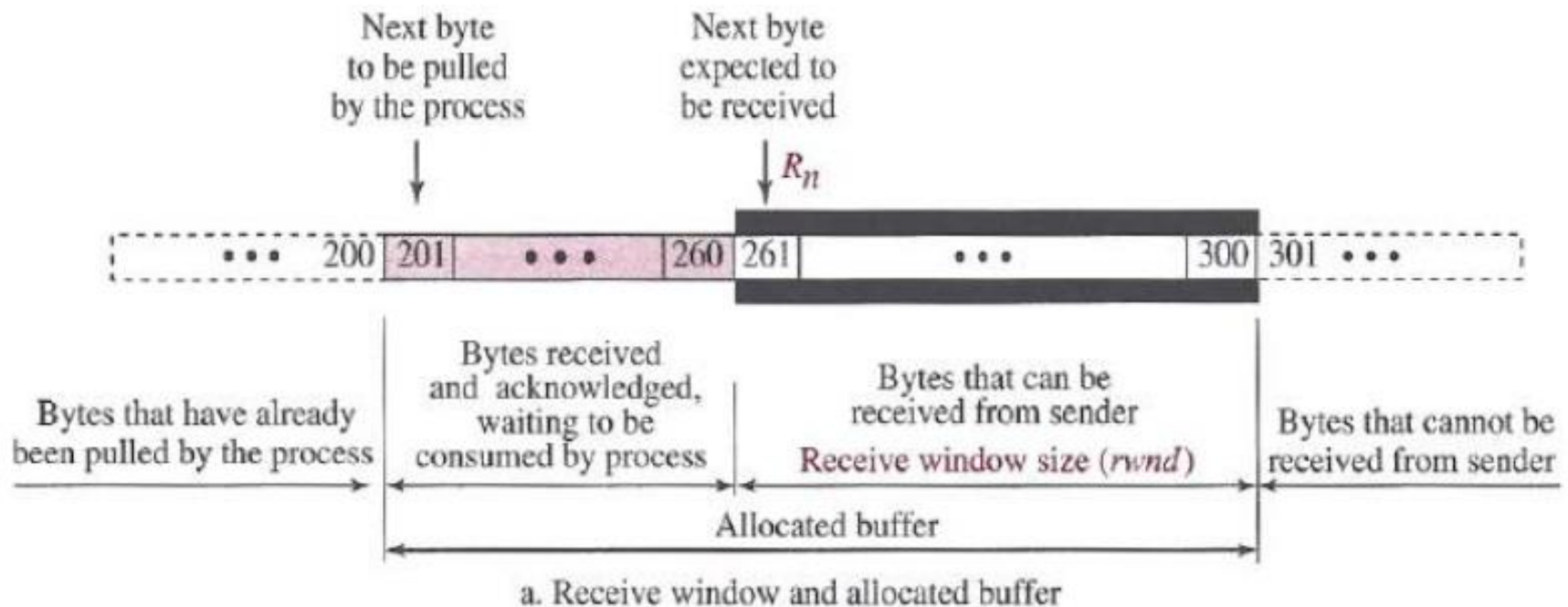


Modified SR (for Send Window)



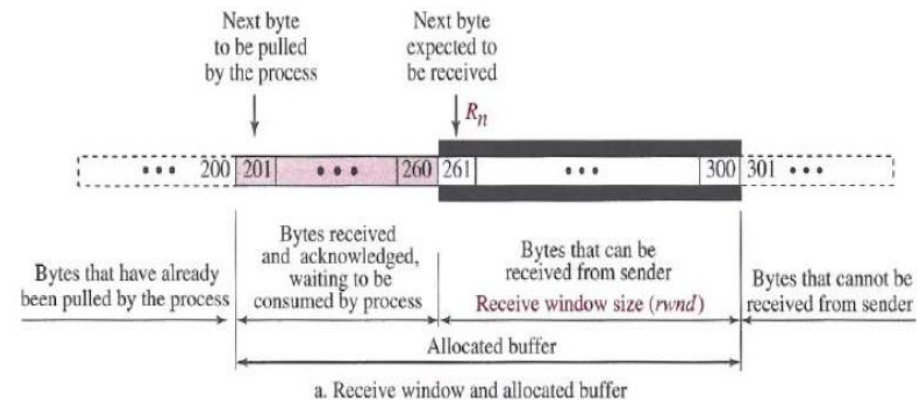
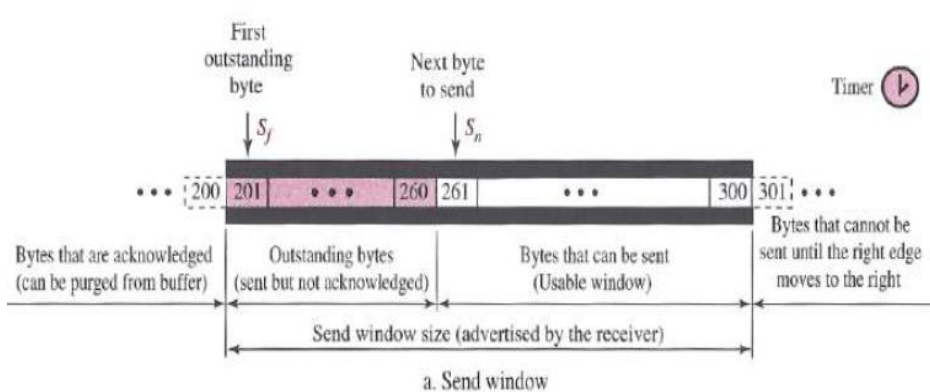
- Sending window in TCP follows Selective-Repeat (SR) protocol with few modifications
 - The window size in SR = number of packets,
 - But, the window size in TCP = number of bytes.
 - SR requires individual ACK of each packet that was sent;
 - But, TCP sends ACK for the next packet that it is expecting
 - SR protocol may use several timers (for general transmission and selected retransmission)
 - But, TCP protocol uses only one timer.
 - Window size can be changed dynamically in TCP

Receive Window in TCP



Modified SR (for Receive Window)

- **Receive window** in TCP is little different than that in SR
 - TCP allows the receiving process to **pull data** at its own pace.
 - The receive window size (***rwnd***) determines the number of bytes that the receive window can accept from the sender before being overwhelmed (**flow control**).
- rwnd* = buffer size - number of bytes waiting to be pulled**
- **ACK** in SR is selective, but ACK in TCP is **cumulative**.
 - Retransmission is selective in SR, but **oldest unACKed** segment is retransmitted in TCP

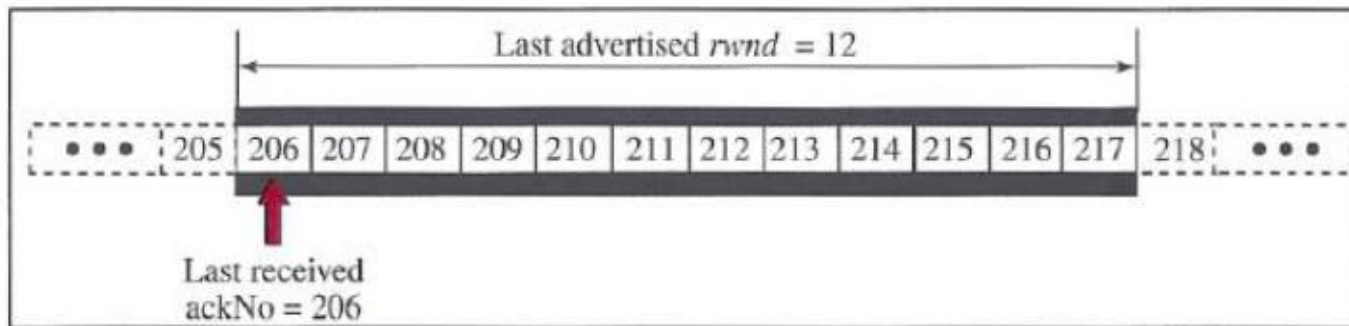




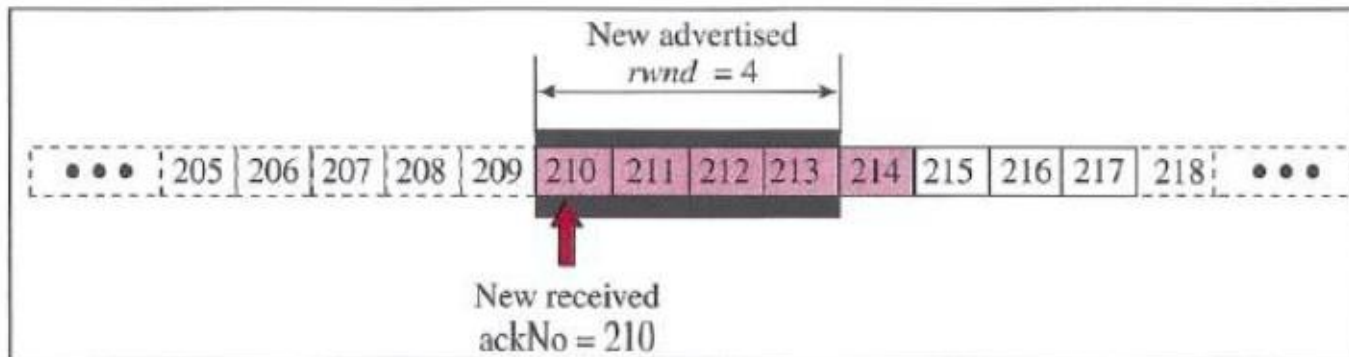
Shrinking of Windows

$\text{new ackNo} + \text{new } rwnd \geq \text{last ackNo} + \text{last } rwnd$

If $rwnd=0$, it instructs for “window shutdown”



a. The window after the last advertisement



b. The window after the new advertisement; window has shrunk

Silly Window Syndrome

- Performance issue occurs when
 - Sending **application** program **creates data slowly**
 - OR, Receiving **application** program **receives data slowly**
 - OR, For **both the above**
- Example:
 - Sending process **generating each byte very slowly**
 - Sending TCP sends many 41 bytes segment (20 byte TCP header + 20 byte IP header + 1 byte data)
- **Two types** to address
 - Syndrome created by sender
 - Syndrome created by receiver

Solution



- **Naïve solution** faces a trade-off optimization
 - If TCP waits too long, it may delay the process
 - If TCP does not wait for long, it may end up sending small segment
- **Better Solution for sender:** Nagle's Algorithm
 - Sending TCP sends the **1st segment** as it is.
 - **2nd segment onwards**, the sending TCP **accumulates data** in sending buffer and **waits until**
 - Either the receiving TCP sends an ACK
 - Or enough data have accumulated **to fill the maximum-size segment** (MSS)
- **Better Solution for receiver:** Clark's two algorithms
- First algorithm:
 - send an ACK as soon as the data arrive,
 - but to **announce a window size of zero** until
 - either there is enough space to accommodate a segment of maximum size
 - or until at least **half of the receive buffer is empty**.
- Second algorithm:
 - delay sending the ACK.
 - The **receiver waits** until there is a decent amount of space in its incoming buffer before ACKing the arrived segments.

Thanks!

Content of this PPT are taken from:

- 1) **Computer Networks: A Top Down Approach**, by J.F. Kurose and K.W. Ross, 6th Eds, 2013, Pearson Education.
- 2) **Data Communications and Networking**, by B. A. Forouzan , 5th Eds, 2012, McGraw-Hill.
- 3) **Chapter 3 : Transport Layer**, PowerPoint slides of “Computer Networking: A Top Down Approach”, 6th Eds, J.F. Kurose, K.W. Ross