### **Exercise 8**

December 15th, 2016



## **TCP** congestion control

- N=200, RTT=200ms, MSS=1000 bytes, sender just sent a complete window!
  - a) Assuming no loss, what is the throughput (in terms of MSS and RTT and in terms of Megabit/s) of this message exchange?

$$throughput = \frac{segments \cdot MSS}{RTT} = \frac{200 \cdot 8000bit}{0.2s} = 8000000 \frac{bit}{s} = 8 \frac{Mbit}{s}$$



### TCP congestion control cont'd

- b) Suppose TCP is in its congestion avoidance phase. Assuming no loss, what is the window size (in terms of segment) after the N = 200 segments are acknowledged?
- From the lecture:
  - When CongWin is above Threshold, sender is in congestion-avoidance phase, window grows linearly.
  - CongWin is in units of MSS



# TCP sender congestion control

State	Event	TCP Sender Action	Commentary
Slow Start (SS)	ACK receipt for previously unacked data	CongWin = CongWin + MSS, If (CongWin > Threshold) set state to "Congestion Avoidance"	Resulting in a doubling of CongWin every RTT
Congestion Avoidance (CA)	ACK receipt for previously unacked data	CongWin = CongWin+MSS * (MSS/CongWin)	Additive increase, resulting in increase of CongWin by 1 MSS every RTT
SS or CA	Loss event detected by triple duplicate ACK	Threshold = CongWin/2, CongWin = Threshold, Set state to "Congestion Avoidance"	Fast recovery, implementing multiplicative decrease. CongWin will not drop below 1 MSS.
SS or CA	Timeout	Threshold = CongWin/2, CongWin = 1 MSS, Set state to "Slow Start"	Enter slow start
SS or CA	Duplicate ACK	Increment duplicate ACK count for segment being acked	CongWin and Threshold not changed



## TCP congestion control cont'd

 b) Suppose TCP is in its congestion avoidance phase. Assuming no loss, what is the window size (in terms of segment) after the N = 200 segments are acknowledged? (CongWin = CW)

$$\circ$$
 in one RTT:  $CW = CW + MSS \cdot \left(\frac{MSS}{CW}\right)$ 

Each ack increases the CW by MSS/CW, which is 8000Bit/200=40Bit. As 200 acks arrive, the window is increased by 8000Bit which is exactly 1MSS, therefore CW=200+1! Note CW = 200+1 MSS!



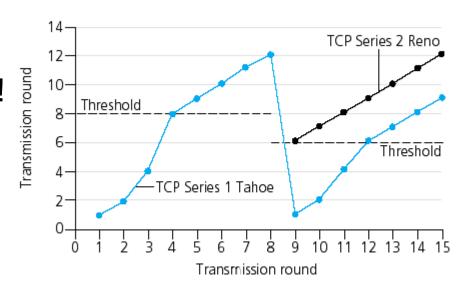
#### **TCP-Reno and Tahoe**

 What is the difference between the two congestion control algorithms TCP-Tahoe and TCP-Reno?



### **TCP-Reno and Tahoe**

 Difference in handling timeouts and triple duplicate acks!



 Tahoe always down to 1MSS, Reno distinguishes: 3 duplicate ACKs-> go down to 50% then CA, timeout -> go down to 1MSS



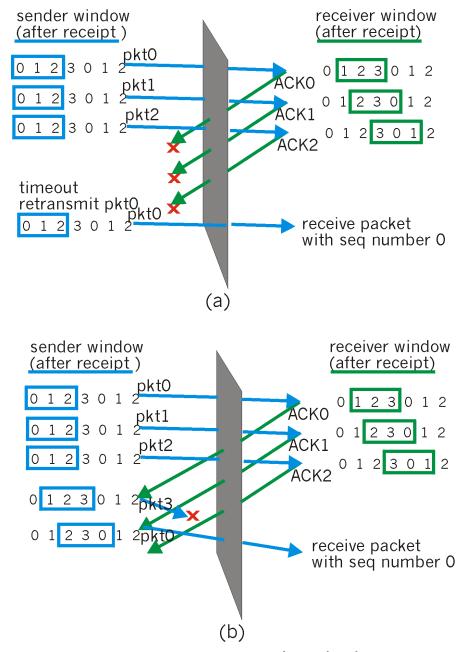
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### **Selective Repeat**

 Please explain the selective repeat dilemma and name a solution to prevent its occurrence.



 Dilemma occurs on a limited sequence range and large window size.
 Solution: Window size should be maximally half of the sequence range!





#### TCP vs. UDP

 Please name at least three differences between UDP and TCP.



#### TCP vs. UDP

- TCP is connection oriented, UDP is not
- TCP is a reliable data transfer protocol, UDP is not reliable
- TCP enables in-order delivery, UDP does not guarantee inorder deliver
- UDP has less overhead (lightweight) compared to TCP (heavy load due to ordering, window maintenance etc...)
- TCP uses flow control, UDP does not
- TCP uses congestion control, UDP does not



### Choosing a protocol

- If you would like to transfer a file, which transport protocol would you use? Which protocol would you use for voice traffic?
  - File: TCP as it is reliable, in-order delivery.
     Receiver can directly pipe data contents into file
  - Voice: UDP as it is lightweight, small in-orders cannot be heard and reliability has no advantage if delivery takes to long



### **TCP** fast retransmit

- Please explain TCP fast retransmit.
  - Time-out period often relatively long:
    - long delay before resending lost packet
  - Detect lost segments via three duplicate ACKs.

Fast retransmit:
resend segment
before timer
expires, directly
after receiving
three duplicate
acks



### Flow vs. congestion control

- What is the difference between flow control and congestion control?
- Flow Control: Prevent overwhelming the receiver by sending too much data. Reduce sending rate if receiver's buffer fills up.
- Congestion Control: React on congestion in the network (on the path to the receiver).
   Reduce sending rates based on congestion observation (deduction by seeing delayed acks, lost acks etc.)



### Estimated vs. sampled RTT

Why is an EstimatedRTT used to calculate the TCP timeout instead of the recently sampled RTT?
RTT: gaia.cs.umass.edu to fantasia.eurecom.fr

- Exponential weighted moving average
- □ influence of past sample decreases exponentially fast

SampleRTT fluctuates
 too much. EstimatedRTT + safety margin is a safer guess to set the timer.

