ECE 356/COMPSI 356 Computer Network Architecture

TCP Congestion Control

Wednesday November 6th, 2019



Recap

- Last lecture:
 - TCP reliable communications
 - ≻TCP flow control
 - TCP connection establishment

• Readings for this lecture: PD 6.3



Lecture Outline

- Understanding congestion
- Principles of congestion control
- Congestion control algorithm components:
 - Slow start
 - Congestion avoidance
 - Fast recovery
- Congestion control as a feedback control system



Recap: Flow Control vs. Congestion Control

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

Congestion control: throttling the sender due to congestion on the network



Understanding Congestion





• Like road congestion, if we could also lose cars in transit



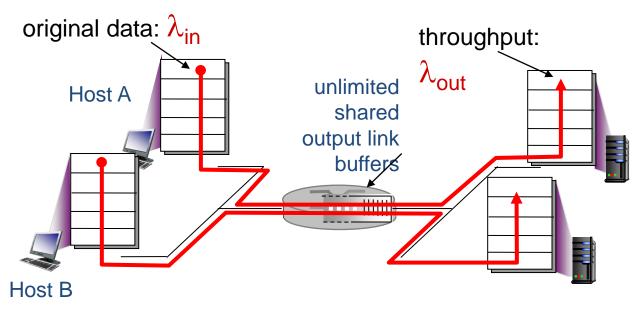
Understanding Congestion

• Informally: "too many sources sending too much data too fast for network to handle"

Different from flow control

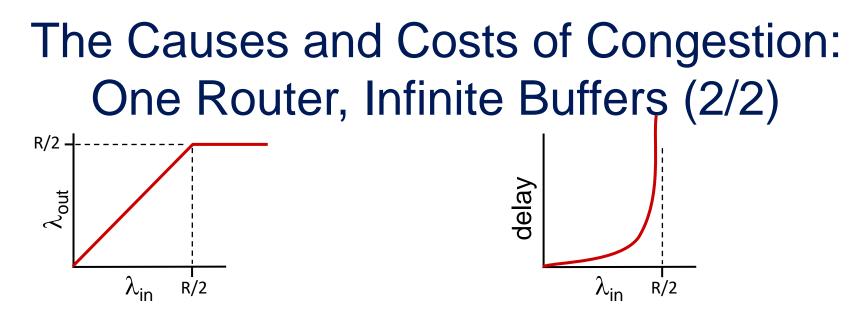
- Manifestations:
 - Long delays (queuing in router buffers)
 - Lost packets (buffer overflow at routers)
- An important and interesting problem
 - Nodes make independent distributed decisions

The Causes and Costs of Congestion: One Router, Infinite Buffers (1/2)



- Two senders, two receivers
- One router, infinite buffers
- Output link
 capacity: R
- *No*

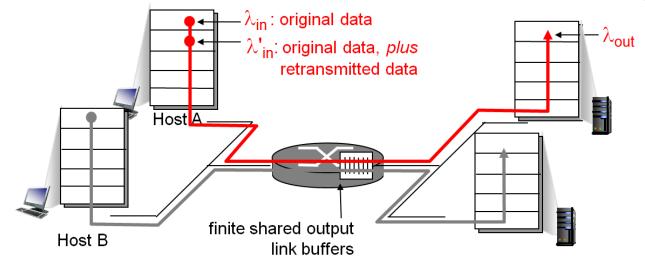
retransmissions



 Maximum per-connection throughput: R/2

- Large delays as arrival rate, λ_{in} , approaches capacity
- Cost of congestion: large queuing delays experienced as packet-arrival rates near link capacity

The Causes and Costs of Congestion: One Router, Finite Buffers (1/3)



Sender retransmission of timed-out packets

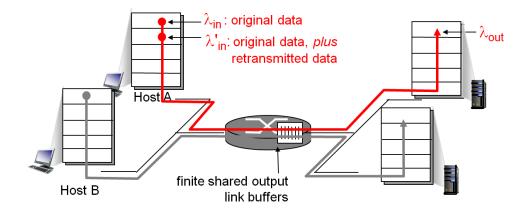
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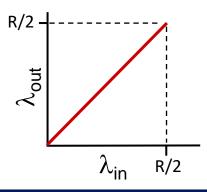
> Application-layer input = application-layer output: $\lambda_{in} = \lambda_{out}$

> Transport-layer input includes *retransmissions* : $\lambda_{in}^{\prime} \geq \lambda_{in}$

The Causes and Costs of Congestion: One Router, Finite Buffers (2/3)

- Idealization: perfect knowledge
 - Sender sends only when router buffers available

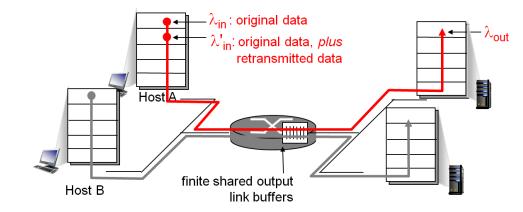


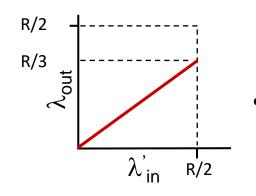


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The Causes and Costs of Congestion: One Router, Finite Buffers (3/3)

 Packets can be lost, dropped at router due to full buffers

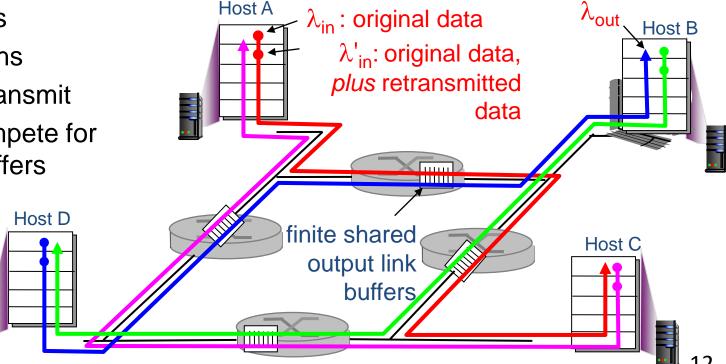




 Cost of congestion: sender must perform retransmissions in order to compensate for dropped packets due to buffer overflow

The Causes and Costs of Congestion: Multi-Hop Paths (1/3)

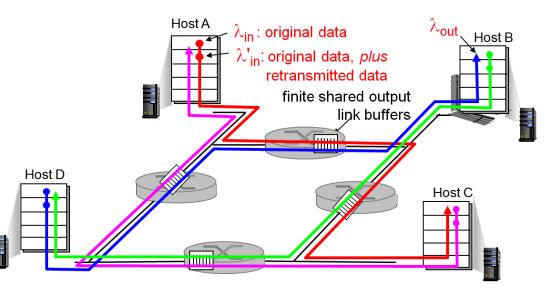
- Four senders
- Multihop paths
- Timeout/retransmit
- Senders compete for space on buffers



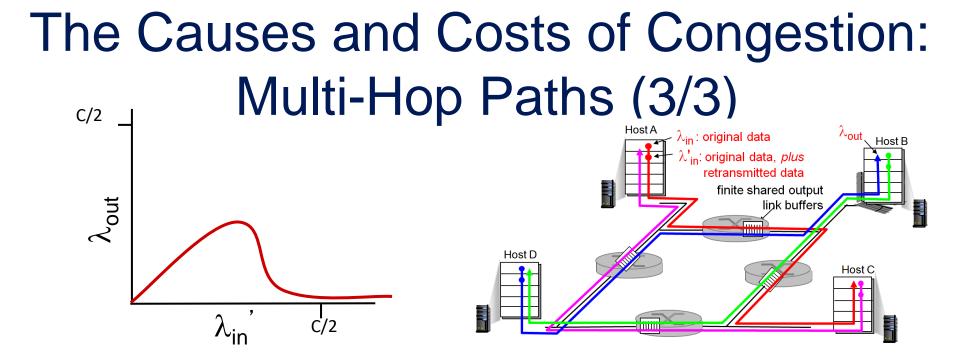


The Causes and Costs of Congestion: Multi-Hop Paths (2/3)

- <u>Q</u>: what happens as λ_{in} and λ_{in} increase?
- A: as red λ_{in} increases, all arriving blue pkts at upper queue are dropped, blue throughput → 0







 Cost of congestion: when a packet is dropped along the path, the capacity used for it is wasted

History (1/2)

- The original TCP/IP design did not include congestion control
 - Receiver uses advertised window to do flow control
 No exponential backoff after a timeout



History (2/2)

- It led to **congestion collapse** in October 1986
 - The NSFnet phase-I backbone dropped three orders of magnitude from its capacity of <u>32 kbit/s</u> to **40 bit/s**
 - 800x difference
 - This continued until end nodes started implementing Van Jacobson's congestion control between 1987 and 1988



Understanding Congestion: Key Points to Remember

- Too many sources sending too much data too fast for network to handle
 - A network-level phenomenon
- Congestion control ≠ flow control
- Costs of congestion include:
 - Large queuing delays
 - Retransmissions to compensate for packets dropped at intermediate routers
 - > Wasted work in forwarding packets that will be dropped

Lecture Outline

- Understanding congestion
- Principles of congestion control
- Congestion control algorithm components:
 - Slow start
 - Congestion avoidance
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- Congestion control as a feedback control system



Congestion Control: Challenge

- Send at the "right" speed
 Fast enough to keep the pipe full
 - But not to overload the network
 - ➤Share nicely with other senders



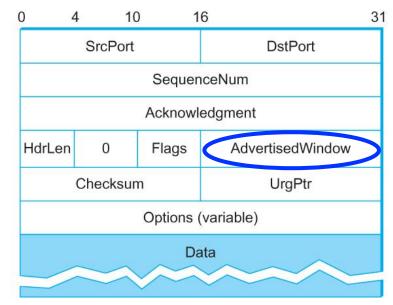
Congestion Control: Approach

- Each sender limits the rate at which it sends traffic into its connection, as a function of *perceived network congestion*
 - ➤ Q: How does the sender limit the rate?
 - > Q: How does the sender perceive congestion?
 - Q: Which algorithm does the sender use to change its send rate?



How Does the Sender Limit Transmission Rate?

- CongestionWindow
- Counterpart to flow control's AdvertisedWindow
 - But, unlike it, is not explicitly signaled
- Maximum number of bytes in transit: *min(CongestionWindow, AdvertisedWindow)*
 - Window-based congestion control



How Does a Sender Perceive Congestion?

- Packet loss is a congestion signal
- Loss events: familiar retransmission triggers
 - Timeout of a retransmission timer
 - *Nothing* is getting through?
 - Receipt of three duplicate ACKs
 - *Something* is passing through the channel



Congestion Detection: Wireless Network Complications

- Recall that wireless networks are much more error prone than wired networks
- In wireless networks, loss ≠ congestion
 ➤ Could be due to weaker signal, interference
 ➤ A large number of packets can get lost
- TCP can slow down to a crawl



TCP for Wireless: Active Area of Research

- One option: splitting the connection into wired and wireless segments
 - Creating a *middlebox*
 - Deviating from end-to-end transport layer architecture
- Another option: distinguish between congestion and bit errors
 - Other congestion clue: explicit congestion notification
 - Another one: increasing RTT values

How Does a Sender Know There is No Congestion?

- Receiving acknowledgements
- Increase congestion window size when acknowledgements are received
 - \succ Acknowledgements arrive slowly \rightarrow slow increase
 - \succ Acknowledgements arrive quickly \rightarrow fast increase
- "Self-clocking" mechanism

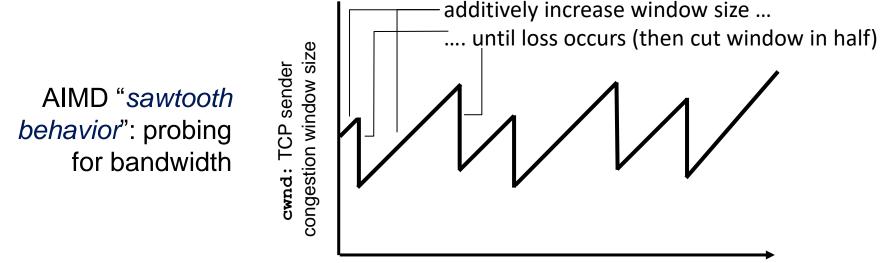


Algorithm: Additive Increase Multiplicative Decrease (1/2)

- Bandwidth probing
- Sender increases transmission rate (window size), probing for usable bandwidth, until loss occurs
 - Additive increase: increase cwnd by 1 MSS every RTT until loss detected
 - > Multiplicative decrease: cut cwnd in half after loss



Algorithm: Additive Increase Multiplicative Decrease (2/2)



Multiple Flavors of TCP

- TCP Tahoe, Reno, Vegas, BBR, CUBIC, ...
- Different feedback signals
- Different specifics of sawtooth patterns



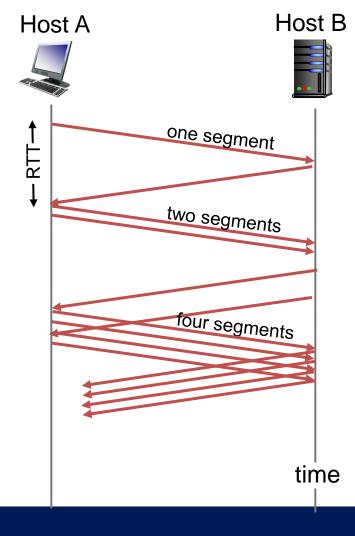
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TCP Slow Start (1/2)

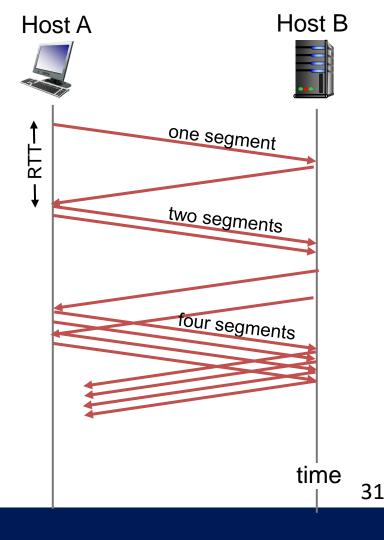
- When connection begins, increase rate exponentially until first loss event
 - Initially cwnd = 1 MSS
 - More in modern TCP variants
 - Double cwnd every RTT





TCP Slow Start (1/2)

- Done by incrementing cwnd for every ACK received
 - Incrementing per ACK, not per segment count
 - Same if acknowledging less than 1 MSS, or many consecutive transmissions
- Summary: initial rate is slow but ramps up exponentially fast



Switching from Slow Start to Congestion Avoidance

Q: when should the exponential increase stop?

Switch to <u>linear increase</u>: **congestion avoidance**

A: when **cwnd** gets to 1/2 of its value before timeout

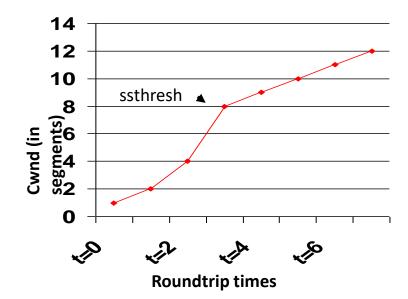
Implementation:

- Variable ssthresh
- On loss event, ssthresh is set to 1/2 of cwnd just before loss event



An Example of Slow Start/Congestion Avoidance

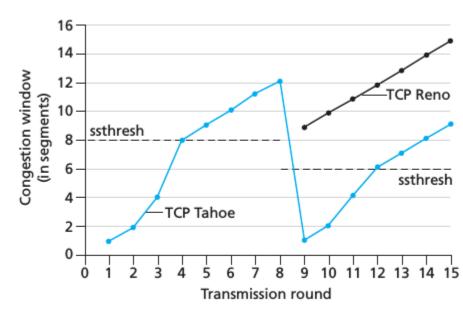
For *ssthresh* = 8 *MSS*





Slow Start: Reacting to Losses

- Timeout
 - > ssthresh \leftarrow cwnd/2
 - \succ cwnd \leftarrow 1 MSS
 - Slow start begins anew
- 3 duplicate ACKs
 - Fast retransmit
 - Enters fast recovery stage

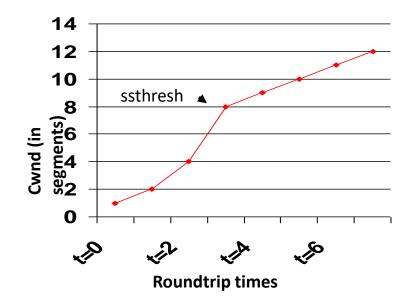


TCP Tahoe example



TCP Congestion Avoidance

- On entry to congestion avoidance stage, cnwd is 1/2 the value of what it was when congestion was last encountered
 - Congestion could be just around the corner
- Conservative growth approach: increase the value of cwnd by 1 MSS every RTT



TCP Congestion Avoidance: Exiting

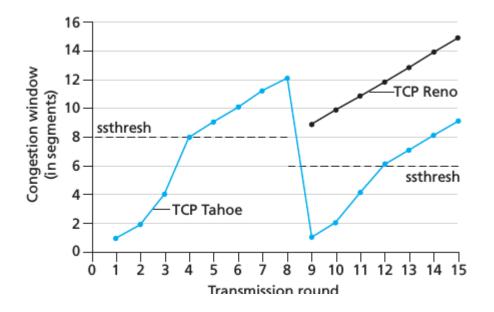
- On a timeout:
 - cwnd set to 1 MSS
 - \succ ssthresh set to 1/2 cwnd when timeout occurred
 - To slow start state
- On a triple duplicate ACK:
 - Fast retransmit
 - \succ cwnd ← cwnd/2 + 3 MSS
 - \succ ssthresh ← cwnd/2
 - ➤ To fast recovery state

TCP Fast Recovery

- Recommended, but not required
- Avoiding slow start
 - The value of cwnd is increased by 1 MSS for every duplicate ACK received for the missing segment that caused TCP to enter fast recovery state
- When ACK arrives for the missing segment:
 - \succ cwnd \leftarrow ssthresh
 - Enter congestion avoidance



Evolution of TCP Congestion Window: Triple Duplicate ACK

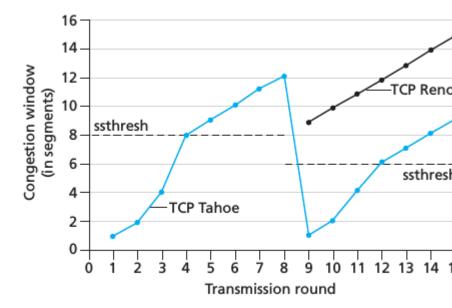


- After a triple duplicate ACK:
 - \succ ssthresh \leftarrow cwnd/2
- TCP Tahoe: $cwnd \leftarrow 1$
- TCP Reno: cwnd ← cwnd/2
 + 3 MSS

• TCP Tahoe: no fast recovery; TCP Reno: fast recovery

Congestion Control Mechanisms: Key Points to Remember

- Congestion window follows a sawtooth pattern
 - Grows first exponentially, then linearly, until a loss event occurs

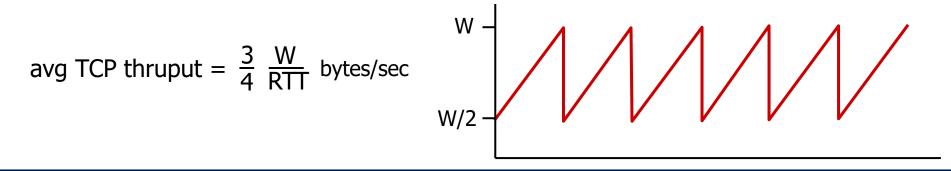




Macroscopic Behavior of TCP

- Avg. TCP throughput as function of window size, RTT?
 > Ignore slow start, assume always data to send
- W: window size (measured in bytes) where loss occurs
 > Avg. window size (# in-flight bytes) is ¾ W
 - > Avg. throughput is 3/4W per RTT

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Congestion control as a feedback control system

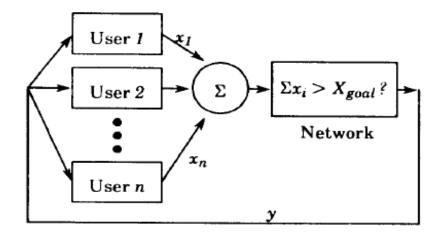


Proof of Optimality

- AIMD was developed based on engineering insight and experimentation
- Ten years after, theoretical analysis showed that the congestion control algorithm is optimal
 - Stable
 - ➤ Fair



Why Does it Work?



- A feedback control system
- The network uses feedback y to adjust users' load $\sum x_i$



Goals of Congestion Avoidance

- Efficiency
- Fairness
- Distributedness
 - A centralized scheme requires complete knowledge of the state of the system
- Convergence
 - > The system approach the goal state from any starting state



Metrics to Measure Convergence

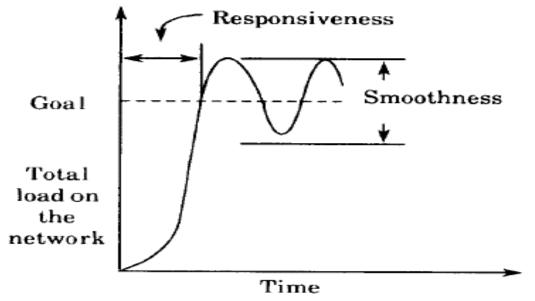
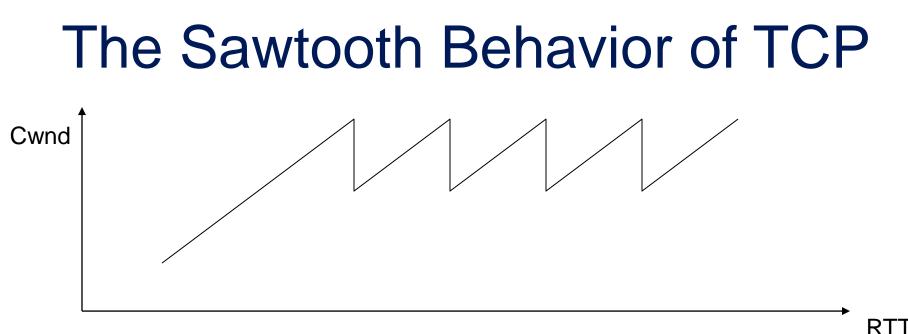


Fig. 3. Responsiveness and smoothness.

- Responsiveness
- Smoothness





- For every ACK received
 > Cwnd += 1/cwnd *MSS
- For every packet lost
 - ➤ Cwnd /= 2

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TCP Congestion Control: Key Points to Remember (1/3)

- Network congestion is problematic. It leads to:
 - Delays
 - Segment losses
 - Wasted work of the network
- TCP employs window-based congestion control
 - Maximum number of bytes in transit: min(CongestionWindow, AdvertisedWindow)
 - Sender *probes the network* by injecting more and more data in it
 - Backs off when encountering losses

TCP Congestion Control: Key Points to Remember (2/3)

AIMD "sawtooth behavior": probing for bandwidth

additively increase window size until loss occurs (then cut window in half) congestion window size TCP sender : pund : time



TCP Congestion Control: Key Points to Remember (3/3)

16 Algorithm component: 14 Slow start: exponential A window ments) 12-TCP Rend growth of cwnd 10sstb Congestion avoidance: linear Cong (in growth of cwdn ssthresh (Recommended) fast TCP Tahoe recovery: avoiding slow start in case of duplicate ACKs Transmission round



Next Lecture

- Network resource allocation
 >Queue management
 - Congestion avoidance

