<u>CSC2209</u> Computer Networks

Congestion Control (end-host view)

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Today's Questions

- How fast should a transmitter send data?
 - Not too slow....
 - Not too fast...
 - Just about right!
- Shouldn't be faster than receiver can process
 - This is called...
- Shouldn't be faster than network can process
 - This is called...

Congestion Control Goals

Congestion Control Goals

- Efficiency
 - Utilize all available bandwidth
- Fairness
 - All hosts get equal access to bandwidth
- Distributed implementation
 - Only require state at endpoints
- Convergence
 - For constant load, arrive at single solution for using/sharing bandwidth

Questions

- How to detect congestion?
- How to limit sending data rate?
- How fast to send?
- How to achieve stability?

Detecting Congestion

Detecting Congestion

- Implicit signaling
 - Packet loss
 - Assumes congestion is primary cause of packet loss
 - Packet delay
 - RTT increases as packets queue
 - Packet inter-arrival time is a function of bottleneck link
 - Pros/cons?
- Explicit signaling
 - Source quench: router sends ICMP "Hey buddy, slow down"
 - ECN: router marks packet on how full queue is
 - Hop-by-hop backpressure

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 - Artificially constrain number of outstanding packets allowed
 - Increase window to send faster, decrease to send slower
 - Pro: cheap to implement, good failure properties
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 - Send x packets in period y
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 - Cons: per connection timers, what if receiver fails

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How fast to send?

- Ideally: keep equilibrium at "knee" of power curve
 - Find "knee" somehow
 - Keep number of packets in flight the same
 - Don't inject new packet until old one left the network
 - What if you guessed wrong?
- Compromise: adaptive approximation
 - If congestion signaled, reduce sending rate by x
 - If data delivered successfully, increase sending rate by y
 - How should x and y be related? Convergence?

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How to achieve stability?

- Additive increase, multiplicative decrease (AIMD)
 - Increase sending rate by a constant (e.g., 1500 bytes)
 - Decrease sending rate by a linear factor (e.g., divide by 2)
- Rough intuition why this works
 - Let L_i be length of queue at time i
 - In steady state: L_i=N, where N is a constant
 - During congestion: $L_i = N + yL_{i-1}$, where y > 0
 - If y is large (close to 1), queue size increases exponentially

Resulting TCP/IP Improvements

- Slow-start
- *Round-trip time variance estimation*
- Exponential retransmit timer backoff
- More aggressive receiver ack policy
- Dynamic window sizing on congestion
- Clamped retransmit backoff (Karn)
- Fast Retransmit

Congestion control means: "Finding places that violate the conservation of packets principle and then fixing them."

>	Packet Conservation
	Principle

Slow-start

- Goal: find equilibrium sending rate
- Quickly increase sending rate until congestion detected
- Algorithm:
 - On new connection, or after timeout, set cwnd=1
 - For each segment acknowledged, cwnd += 1
 - If timeout then cwnd /= 2, set ssthresh = cwnd
 - If cwnd >= ssthresh then exit slow start
- Very confusing name

Adaptive Timing

- How long should we wait for a packet's acknowledgement
 - Too short: spurious timeouts and retransmissions
 - Too long: wasteful
- Old TCP
 - Maintain weighted average of RTT samples: R
 - Timeout set to B*R, where B=2
 - Under high load, this scheme doesn't reflect variation
- Jacobson's contributions
 - Estimate variation, B based on some samples
 - After loss, increase timeout exponentially (by 2)

Fast Retransmit & Fast Recovery

• Fast retransmit

- Timeouts are slow (1 second is shortest TCP timeout)
- When packet is lost, receiver still acks last in-order packet
- Use 3 duplicate ACKs to indicate loss
 - Why 3? When wouldn't this work?
- Fast recovery
 - If ACKs are still arriving, then no need for slow start
 - Divide cwnd by 2 after fast retransmit
 - Increment cwnd by 1 for each duplicate ACK

A TCP Taxonomy

- TCP Tahoe (1988)
 - Slow-start, fast retransmit, congestion avoidance
- TCP Reno (1990)
 - Tahoe + fast recovery
- TCP New-Reno (1996)
 - Reno + partial ACKs
- SACK TCP (1996)
 - Selective acknowledgements
- TCP Vegas (1993)
 - Uses RTT variation to measure congestion
- TCP BIC (2004)
 - Binary search between W_{min} and W_{max}

Congestion Control in POTS

• How is it done?

Congestion Control in POTS

- How is it done?
 - Pricing
 - Call doesn't get through (admission control)

Short Connections

- How are short connections affected by slow-start?
 - What happens if drop in slow-start?
 - What happens when SYN dropped?
- Bottom line: which packet gets dropped matters a lot!
- Are most flows long or short?

Cooperation

- TCP is designed around the premise of cooperation
 What if receiver lies about receiving packets?
- Does over-provisioning always help?
 - Let's look at router queues