CPE348: Introduction to Computer Networks

Lecture #17: Chapter 6



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TCP Congestion Control Overview

- Early on, we talked about:
 - TCP sliding window protocol;
 - TCP adaptive transmission algorithms.
- In this lecture, we will talk about:
 - TCP congestion control protocols
 - Additive Increase/Multiplicative Decrease
 - Slow Start
 - Fast Retransmit and Fast Recovery



TCP Congestion Control – history

- It was introduced into the Internet in the late 1980s by Van Jacobson, ~ 8 yrs after the TCP/IP had become operational.
- Immediately preceding this time, the Internet was suffering from congestion collapse —
 - hosts would send their packets into the Internet as fast as the advertised window would allow
 - congestion would occur at some router, causing pkts drop
 - hosts would time out and retransmit their packets, resulting in even more congestion



TCP Congestion Control – history

Basic idea

- Source determines how much capacity is available in the network,
 - it uses the arrival of an ACK as a signal that one of its packets has left the network,
 - source can then insert a new packet into the network.
- By using ACKs to pace the transmission of packets,
 TCP is said to be self-clocking.



TCP Congestion Control

- Three popular TCP congestion control protocols
 - Additive Increase/Multiplicative Decrease
 - Slow Start
 - Fast Retransmit and Fast Recovery





- Source maintains a new variable CongestionWindow
- TCP's effective window is revised as follows:
 - MaxWindow = MIN(CongestionWindow, AdvertisedWindow)
 - EffectiveWindow = MaxWindow - (LastByteSent - LastByteAcked).
- That is, MaxWindow replaces AdvertisedWindow;
- Source is allowed to send no faster than the slowest component—the network or the destination host—can accommodate.



Minimum

- Question is how source comes to learn an appropriate value for CongestionWindow.
 - First, source sets it as a default value based its perception to the network.
 - Then,
 - Decrease the CongestionWindow when the level of congestion goes up and
 - Increase the CongestionWindow when the level of congestion goes down.
 - Taken together, the mechanism is commonly called additive increase/multiplicative decrease (AIMD)



- Th next question is how source comes to learn congestion occurs.
 - Observation: when packets are not delivered, timeout occurs
 - Then, source interprets timeouts as a sign of congestion and reduces the rate at which it is transmitting.
 - When a timeout occurs, the source sets CongestionWindow to half of its previous value multiplicative decrease
 - When no timeout occurs, the source increments CongestionWindow by 1 additive increase.



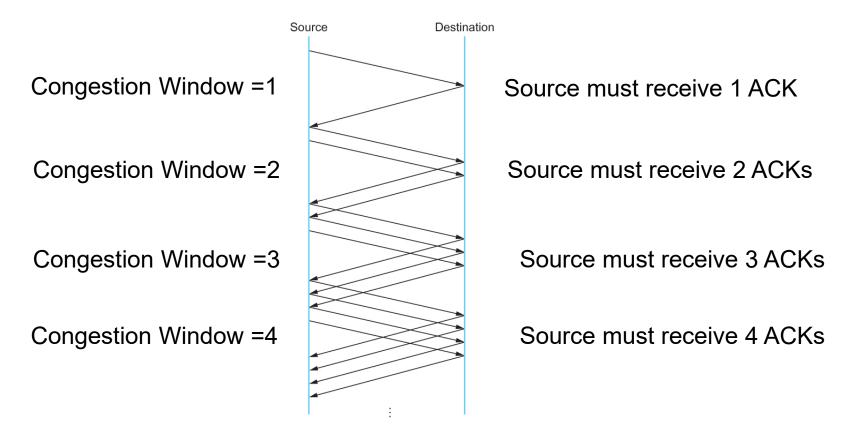
For example

- CongestionWindow is defined in terms of bytes;
- Then,
 - Firstly, CongestionWindow is currently set to 16 packets. If a loss is detected, CongestionWindow is set to 8 packets.
 - Additional losses cause CongestionWindow to be reduced to 4, then 2, and finally to 1 packet.
 - CongestionWindow is not allowed to fall below the size of a single packet, or in TCP terminology, the maximum segment size (MSS).





Another example



Packets in transit during additive increase, with one packet being added each RTT.

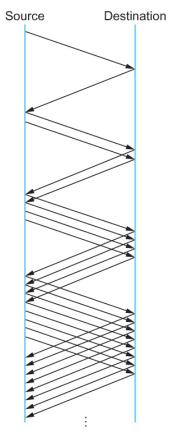


- Drawback of AIMD: too conservative! It takes too long to ramp up a connection starting from scratch.
- TCP therefore provides a second mechanism, ironically called slow start,
 - Increase the CongestionWindow rapidly from a cold start when the CongestionWindow size starts at 1;
 - Slow start effectively increases the congestion window exponentially, rather than linearly (additive increase).

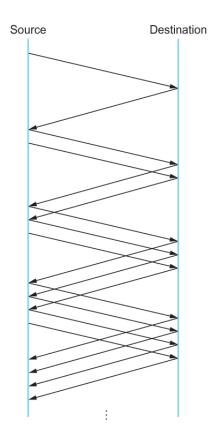


- For example
 - Source starts out by setting CongestionWindow to 1 packet.
 - When the ACK for this packet arrives, source adds 1 to CongestionWindow and then sends 2 packets.
 - Upon receiving the corresponding 2 ACKs, source increments CongestionWindow by 2 and next sends 4 packets.
 - The result is that source doubles the number of packets it has in transit every RTT.





Packets in transit during slow start.



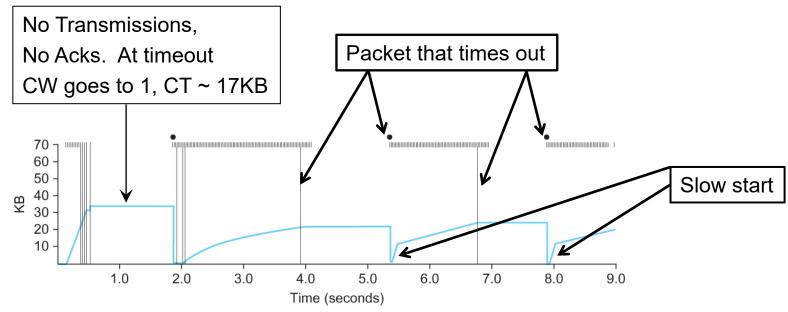
Packets in transit during additive increase



- There are two situations in which slow start is favorable:
 - The first is at the very beginning of a connection:
 - Slow start continues to double CongestionWindow each RTT
 - After a packet is lost, a timeout causes multiplicative decrease to divide CongestionWindow by 2.
 - The second happens when the connection goes dead while waiting for a timeout to occur:
 - After communications for a while, source has a useful value of CongestionWindow, which is called CongestionThreshold;
 - Use CongestionThreshold as the "target" CongestionWindow.
 - Slow start is used to rapidly increase the sending rate up to this value, and then additive increase is used beyond this point. (Combination! Fine-resolution!)



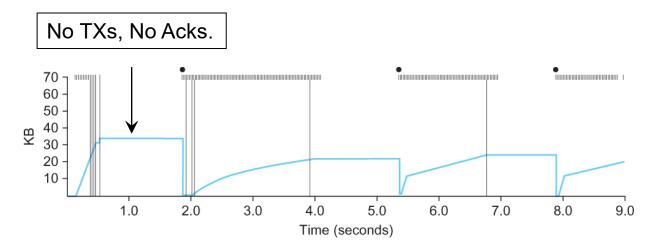
Cont' – Example for timeout occurs



Behavior of TCP congestion control. Colored line = value of CongestionWindow over time; solid bullets at top of graph = timeouts; hash marks at top of graph = time when each packet is transmitted; vertical bars = time when a packet that was eventually retransmitted was first transmitted.



 Issue of slow start + AIMD: TCP timeouts led to long periods of time during which the connection went dead while waiting for a timer to expire.



- A new mechanism called fast retransmit was added to TCP.
- It is a heuristic that sometimes triggers the retransmission of a dropped packet sooner than the timeout.



Basic idea

- Receiver responds with an ACK every time a packet arrives - even if it has already been ACKed
- When an out of order packet is received, TCP resends the same ACK as last time
- TCP cannot ACK the data contained in an out of order packet

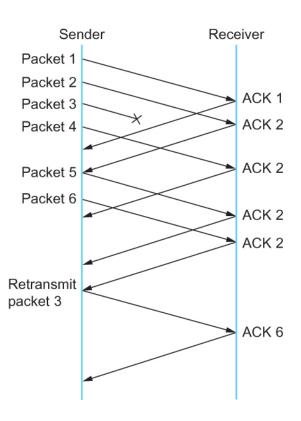
Receiver Assisted Approach



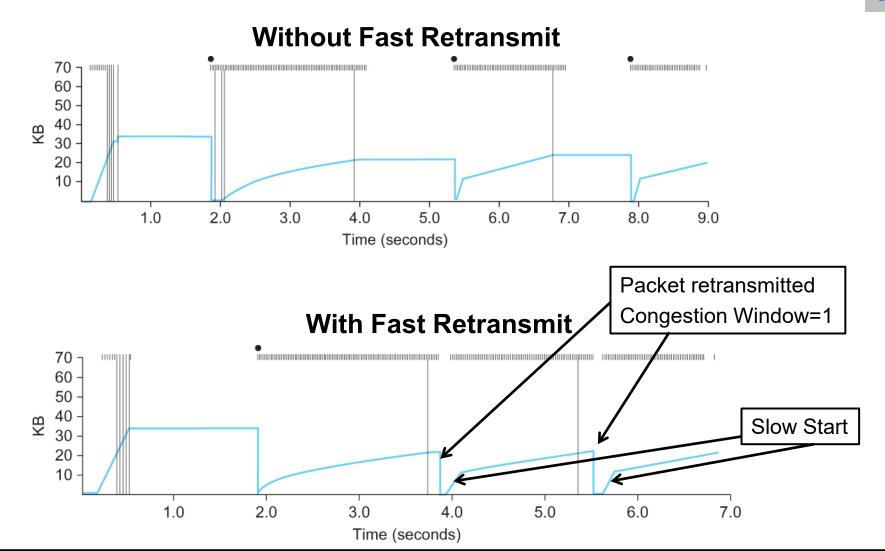
- Basic idea cont'
 - When the source sees a duplicate ACK,
 - It knows that the other side must have received a packet out of order,
 - Suggests that an earlier packet might have been lost.
 - In practice, TCP waits until it has seen three duplicate ACKs before retransmitting the packet.
 - in case the packet is delayed instead of lost



Example









Fast Recovery

- When the fast retransmit mechanism signals congestion,
 - NOT drop the CongestionWindow back to 1 packet and run slow start,
 - it is possible to use the ACKs that are still in the pipe to clock the sending of packets.
- This mechanism, which is called fast recovery.



TCP Congestion Control Summary

- In this lecture, we will talked about:
 - TCP congestion control protocols;
 - Additive Increase/Multiplicative Decrease
 - Slow Start
 - Fast Retransmit and Fast Recovery
 - The (dis-)advantages of them;
 - You are expected to draw the tx diagram given a specific TCP congestion control protocol;

