TCP, Lecture 4

Internet Protocols

CSC / ECE 573

Fall, 2005

N. C. State University

Today's Lecture

- Congestion Control: Fast Retransmit and Recovery
- II. Silly Windows
- III. Urgent Data
- IV. (Some) TCP Options
- V. Router Queue Management

FAST RETRANSMIT AND RECOVERY

Fast Retransmit and Recovery (RFC 2581)

- When an out-of-order segment arrives at the receiver...
 - receiver will generate an ACK with the same sequence number as the previous ACK; called a duplicate ACK
 - indicates that some data is getting through to receiver
- Receipt of 3 duplicate ACKs in a row for segment j is a strong indication that segment j+1 was lost
 - immediately retransmit without waiting for timer to expire: fast retransmit
 - then enter congestion avoidance phase directly (i.e., bypass slow start): fast recovery

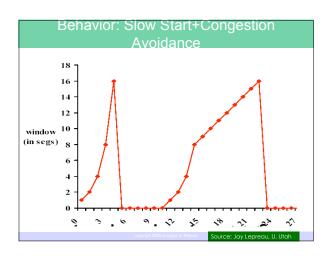
A sends S1 B acks S1 A sends S2 (lost in transmission) A sends S3 B acks S1 A sends S4 B acks S1	
A sends S5 B acks S1 Segment S1, means segme was lost	

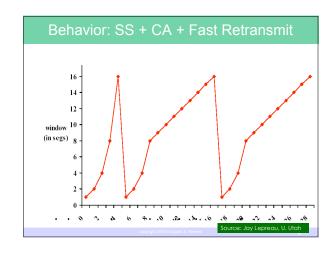
Fast Retransmission and Recovery

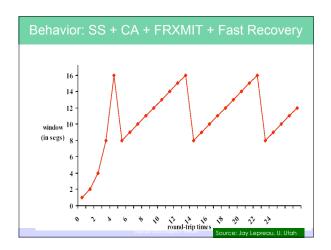
- Sender actions (after receiving 3rd duplicate ACK)
 - retransmit: retransmit segment j+1 without waiting for timeout
 - recover:

 $ssthresh \leftarrow MAX(2, \frac{1}{2} * cwnd)$ $cwnd \leftarrow ssthresh + 3$

- Each time another duplicate ACK arrives...
 - cwnd ← cwnd + 1 /* haven't started congestion avoidance */
 - transmit another packet (if allowed)
- When ACK for a retransmitted segment arrives...
 - $\mathtt{cwnd} \leftarrow \mathtt{ssthresh}$ /* now in congestion avoidance */









"Silly Window" Syndrome (RFC 813)

- A serious problem in sliding window operation
- Causes
 - 1. sending application program creates data slowly
 - 2. receiving application program consumes data slowly
- In either case, data may be sent in small segments
 - inefficient use of bandwidth
 - increased processing by TCP

"Silly Windows" Caused by Receiver

Receiver Buffer is full
Receiving application processes one byte of data

to other end of connection
Sends window update (ACK) allowing one more byte to be sent

New data (one byte) arrives

"Silly Window" Solution

- Clark's solution: do not send window advertisements for 1 byte
- Instead, advertise a window size of zero and wait until
 - 1. there is space for a maximum sized segment of data, or
 - 2. the receive buffer is half-empty
- · Then advertise this new Window Size

Combining Solutions

- Nagle's algorithm: sender accumulates data until "enough" data to send
- Clark's solution: receiver consumes data until "enough" space available to advertise
- These solutions are complementary and can be used together

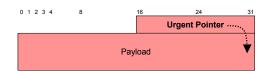
URGENT AND PUSH FLAGS

Urgent Data

- TCP does not provide a separate "control" channel for applications
- · Examples of control
 - ftp: "stop sending data"
 - telnet: interrupt running process, or suspend the telnet session
- · Choices
 - use a second (companion) TCP connection for control, or...
 - insert control into data channel and mark as urgent

Jrgent Data (cont'd)

- Marking urgency: set the URG Flag and set Urgent Pointer to indicate location of the control information
- TCP notifies application "urgent data" has been received
 - processing is application specific



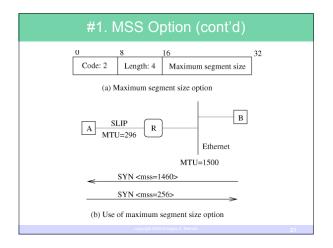
The PSH Flag

- Purposes
 - Sender: forces TCP to send a segment without waiting for further data to be generated
 - Receiver: forces TCP to notify the application that data is waiting to be processed
- Example: after each command typed, during an interactive application
- Prevents TCP buffering (for efficiency) from adding undesirable delivery latency

(SOME) TCP OPTIONS

#1. MSS Option (cont'd)

- MSS = Maximum Segment Size
 - defines the maximum segment size the receiver is willing to accept
 - MSS must be ≤ receiver interface MTU 40 bytes
- Declared during connection establishment phase (i.e., in SYN segments)
 - cannot be specified or changed during data transfer



#2. Window Scale Option (RFC 1323)

- Reminder: optimal Window Size = RTT * receiver bandwidth
- For RTT = 100ms, and bandwidth > 640 KB/s, optimal value is larger than maximum Window Size (64KB)
- Solution: negotiate (during connection establishment only) a scale factor that increases possible window sizes
 - may have different scale factors in the two directions

#2. Window Scale Option (cont'd)

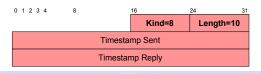
- Scale factor is the exponent of a power-of-two increase in the window size
 - "effective" window size = Window Size × 2^{window-scale-factor}
- Maximum value = 14
 - max effective window size = $(2^{16}-1)^*2^{14}$ ($\approx 2^{30}$)
 - At RTT=100ms, max receiver bandwidth = 10 GB/s

0 1 2 3 4 8 16 24 31

Kind=3 Length=3 Window Scale Factor

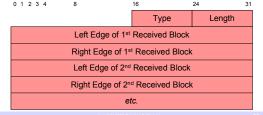
#3. Timestamp Option

- Without timestamps, RTT (in many implementations) is calculated once every window
 - OK for small (e.g., < 8 segment) windows
 - but larger windows require better RTT calculations
- Sender puts timestamp option in segment; option is "reflected" by receiver in the acknowledgment
 - sender can compute RTT for each received ACK



#4. Selective Acknowledgments Option (RFC 2018)

- Selective Acknowledgments: indicate specifically what non-contiguous blocks of data have been received
- · Option format:



#4. Selective Acknowledgments (cont'd

- Receiver notifies sender that non-contiguous blocks of data have been received
 - at most 4 blocks can be specified
 - should be included in all ACK=1 segments that do not acknowledge the highest contiguous sequence number received.
 - report the most recent non-contiguous blocks
- Sender will not retransmit selectivelyacknowledged segments

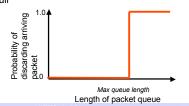
#4. Selective Acknowledgments (cont'd)

- Can result in better throughput when losses are common
- Requires negotiating "SACK-permission" during connection establishment

ROUTER QUEUE MANAGEMENT

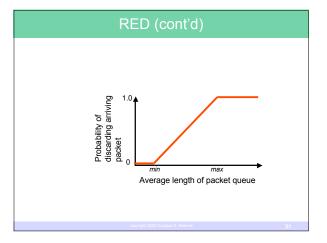
Packet Dropping Policies at Routers

- Incoming packets at a router, after the forwarding decision, are queued for output
- · During congestion, something has to give!
 - "Drop Tail" policy: drop an incoming packet if the queue is already full



RED (RFC 2309)

- Another way: drop the arriving packet randomly, with probability derived from the queue length: RED (random early discard) policy
 - queue length used is an exp. weighted moving average
- · Parameters
 - min and max thresholds per output queue



RFD Evaluation

- · RED considerably more complex than drop-tail
- Claim: RED results in shorter average queue lengths (thus, lower latency)
- · Drop-tail synchronizes losses across flows
 - i.e., they all congest at same time, all back off at same time, then all congest at same time, ...
- · Claim: drop-tail unfair to bursty traffic flows
- · Claim: RED gives significantly better throughput

Summary

- rast Retransmit and Recovery provide improved "steady state" behavior
- Silly" Windows leads to inefficient data transfer
- ddeas about congestion control improvements are never-ending ©
- TCP options provide useful extensions; MSS is universally used

Next Lecture

• The Sockets Network Programming API