

Announcements

- Reading
 - Today: Chapter 6 (6.6)
- Be sure to get the newer version of the net-config module

TCP Congestion Control

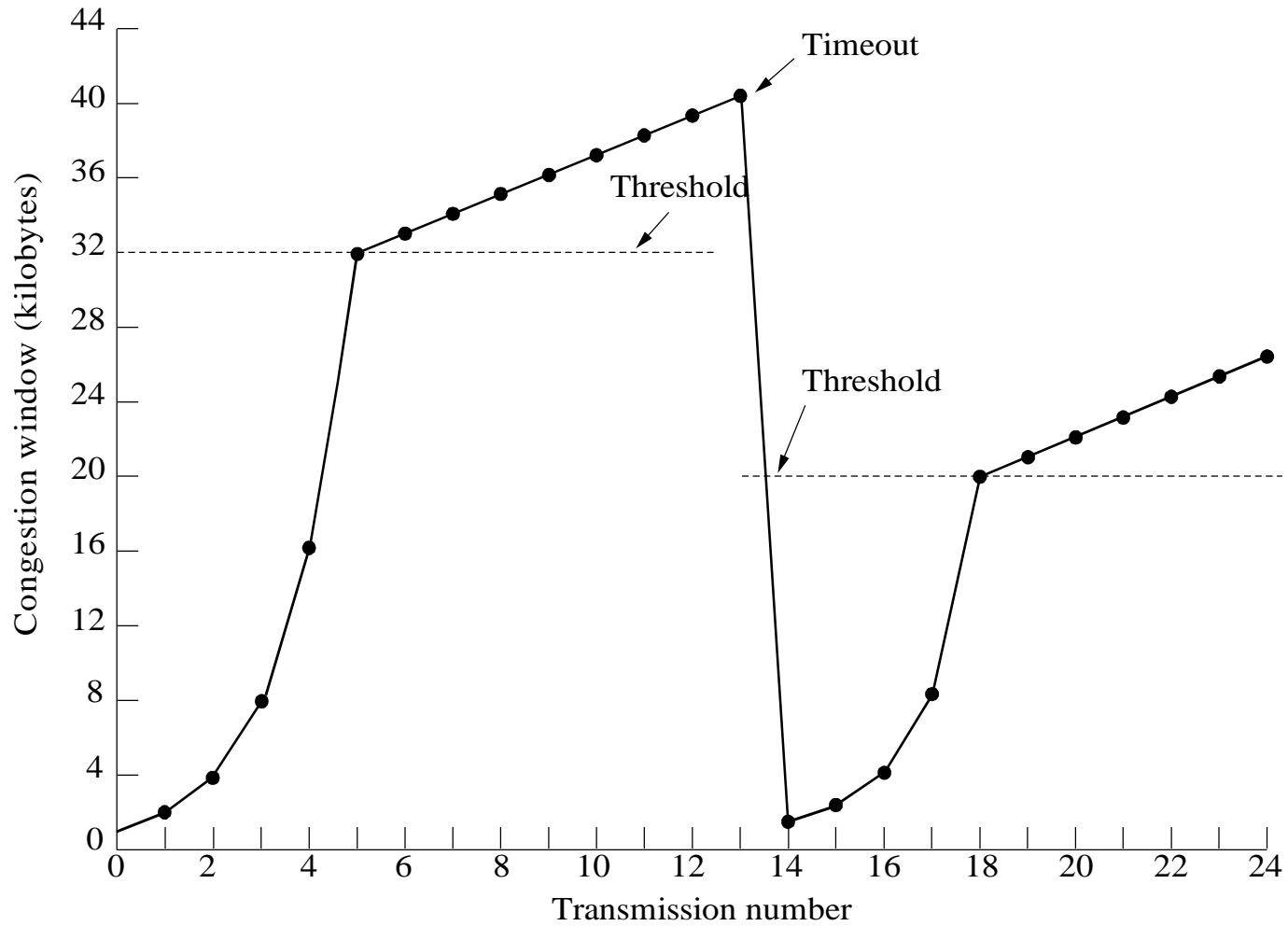
- Detecting Congestion

- In general it is difficult
- But, consider why a packet might be dropped
 - link error - but links are very reliable now
 - buffer overflow --> congestion
- Use re-transmission timeouts as an estimate of congestion

- Dealing with Congestion

- add a second window (congestion window)
 - limit transmissions to $\min(\text{recv window}, \text{congestion window})$
- start with congestion window = max segment window
 - initial max segment is one kilo-byte
 - on a ACK without a timeout
 - if $\text{window} < \text{threshold}$, increment by one max segment
 - otherwise increment by initial max segment
- on timeout
 - cut threshold in half
 - set window size to initial max segment

TCP Congestion Window



From: *Computer Networks*, 3rd Ed. by Andrew S. Tanenbaum, (c)1996 Prentice Hall.

TCP Timer Management

- Problem: How to pick timeout value?
 - need to estimate round-trip latency
 - need low variance in round trip latency
- Solution: dynamic estimates of RTT
 - $RTT = \alpha RTT + (1 - \alpha) M$
M time of an ACK
 $\alpha = 7/8$
 - Need to pick retransmission time
 - old policy, use Timeout = RTT β , with $\beta = 2$
 - estimate standard deviation of RTT using mean deviation
$$D = a D + (1 - a) | RTT - M |$$
$$\text{Timeout} = RTT + 4 * D$$
 - How to update RTT on retransmission's
 - double Timeout on a retransmission

Other TCP Timers

- Persistence Timer
 - Prevents deadlock due to dropped window packets
 - This is a problem if the window is set to 0
- Keepalive Timer
 - Prevents half dead connections
 - may consume bandwidth
 - may kill live connections when net hiccups
- TIMED Wait
 - prevents re-use of a connection before max packet life is over
 - set to twice max packet lifetime

Performance Issues

- **Broadcast storms**
 - response to a broadcast packet sent by many hosts
 - caused by:
 - bad parameter resulting in an error message
 - asking a question everyone has the answer to
- **Reboot storms**
 - RARP queries
 - file servers responding to page requests
- **Delay-bandwidth product**
 - need to buffer at least as many bytes as can be “in flight”
- **Jitter**
 - keep standard deviation of packet arrivals low
 - important for continuous media traffic

How to Measure Performance

- **Ensure sample size is large**
 - repeat experiments for several iterations
- **Make sure samples are representative**
 - consider time of day, location, day of week, etc.
- **Watch for clock resolution/accuracy**
 - don't use two clocks at opposite ends of the network
 - if the clock resolution is poor, aggregate over multiple iterations
- **Know what you are measuring**
 - is a cache going to distort results?
 - is the hardware, OS, device driver, compiler the same?
- **Careful not to extrapolate too far**
 - results generally hold for an operating region, not all values