

CSEP 561 – Retransmissions

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Retransmissions

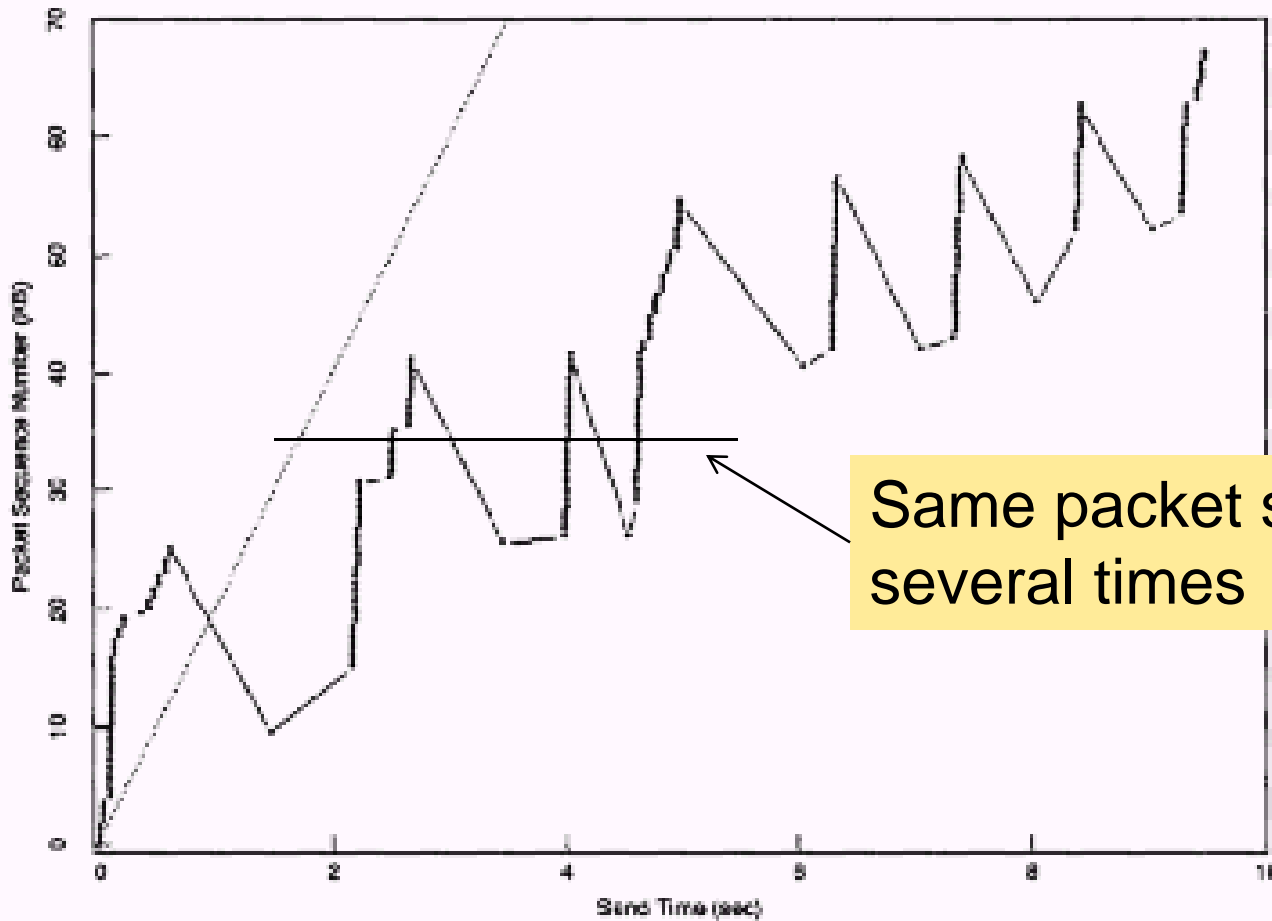
- Focus
 - How do we decide when to retransmit?
- Topics
 - RTT estimation
 - Karn/Partridge algorithm
 - Jacobson/Karels algorithm

Application
Transport
Network
Link
Physical

Deciding When to Retransmit

- How do you know when a packet has been lost?
 - Ultimately sender uses timers to decide when to retransmit
- But how long should the timer be?
 - Too long: inefficient (large delays, poor use of bandwidth)
 - Too short: may retransmit unnecessarily (causing extra traffic)
 - A good retransmission timer is important for good performance
- Right timer is based on the round trip time (RTT)
 - Easy for LANs, varies greatly for wide area (path length, queuing)

Effects of Early Retransmissions



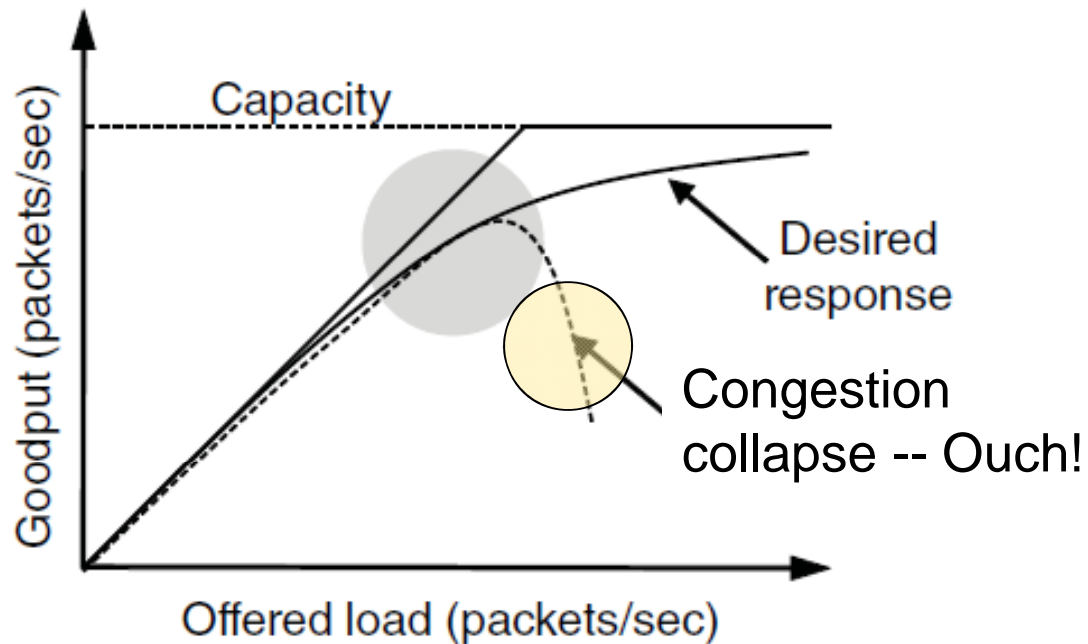
Congestion Collapse

- In the limit, early retransmissions lead to congestion collapse
 - Sending more packets into the network when it is overloaded exacerbates the problem of congestion
 - Network stays busy but very little useful work is being done
- This happened in real life ~1987
 - Led to Van Jacobson's TCP algorithms, which form the basis of congestion control in the Internet today

[See "Congestion Avoidance and Control", SIGCOMM'88]

Congestion Collapse

- Retransmissions lead to throughput, but not goodput

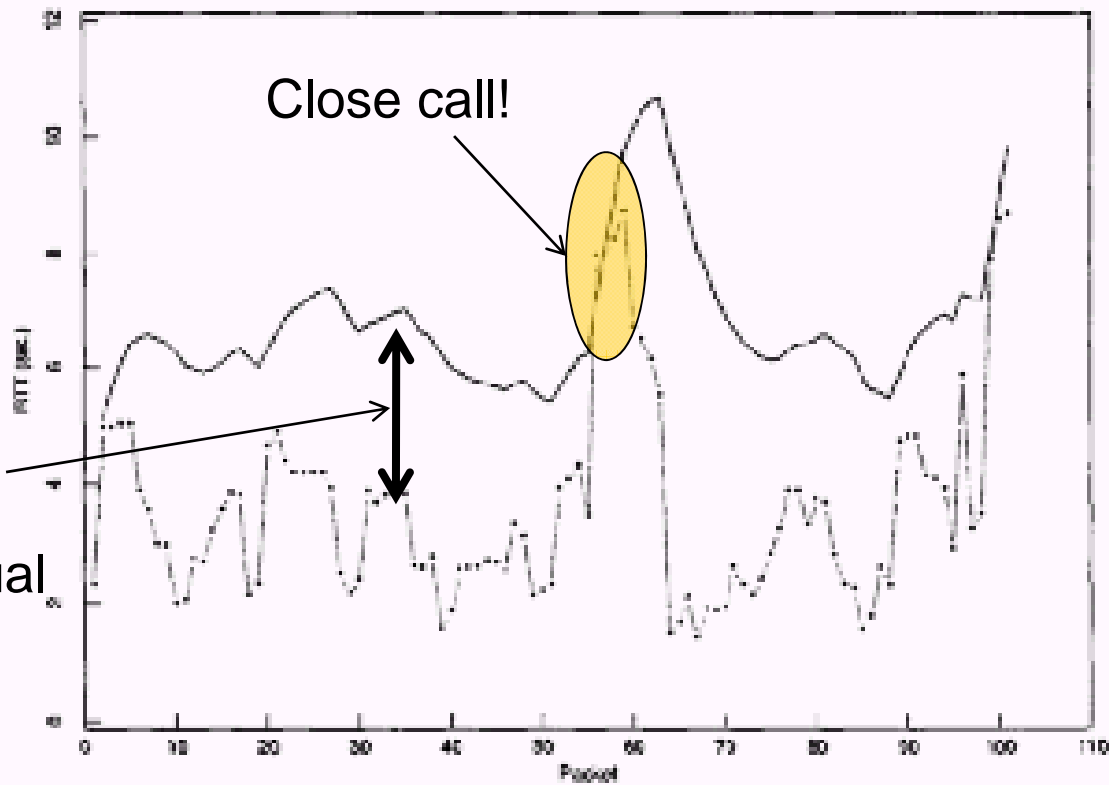


Estimating RTTs

- Idea: Adapt based on recent past measurements
- Simple algorithm:
 - For each packet, note time sent and time ack received
 - Compute RTT samples and average recent samples for timeout
 - $\text{EstimatedRTT} = \alpha \times \text{EstimatedRTT} + (1 - \alpha) \times \text{SampleRTT}$
 - This is an exponentially-weighted moving average (low pass filter) that smoothes the samples. Typically, $\alpha = 0.8$ to 0.9 .
 - Set timeout to small multiple (2) of the estimate

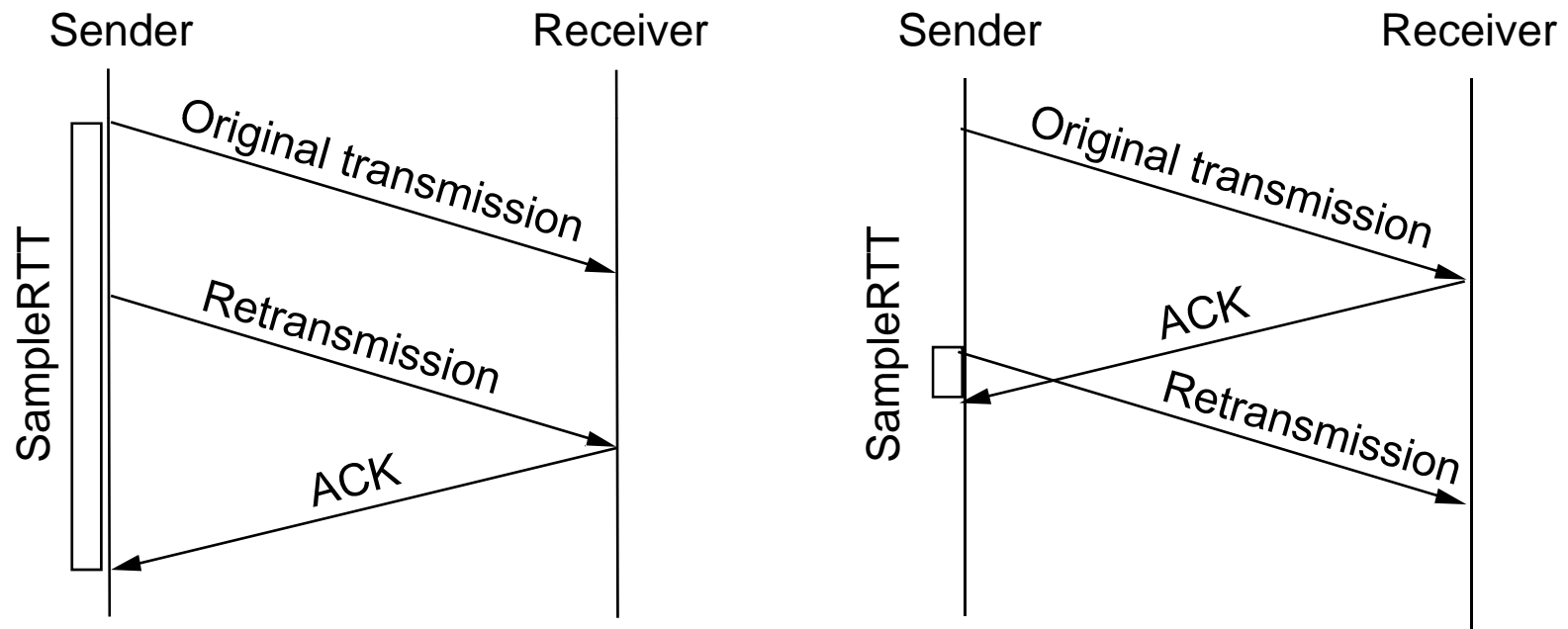
Estimated Retransmit Timer

But has large gap to keep estimate > actual



Karn/Partridge Algorithm

- Problem: RTT for retransmitted packets ambiguous



- Solution: Don't measure RTT for retransmitted packets and do not relax backed-off timeout until valid RTT measurements

Jacobson/Karels Algorithm

- Problem:
 - Variance in RTTs gets large as network gets loaded
 - So an average RTT isn't a good predictor when we need it most
- Solution: Track variance too.
 - $\text{Difference} = \text{SampleRTT} - \text{EstimatedRTT}$
 - $\text{EstimatedRTT} = \text{EstimatedRTT} + (\delta \times \text{Difference})$
 - $\text{Deviation} = \text{Deviation} + \delta(|\text{Difference}| - \text{Deviation})$

 - $\text{Timeout} = \mu \times \text{EstimatedRTT} + \phi \times \text{Deviation}$
 - In practice, $\delta = 1/8$, $\mu = 1$ and $\phi = 4$

Estimate with Mean + Variance

