

Advanced Computer Networks

Congestion Control over Large Bandwidth-Delay Product Networks

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Summer 2007

Review: TCP congestion control

- Loss-based
 - how to detect/react to packet losses
- Delay-based
 - how to react to delay changes
- Rate-based
 - how to determine the TCP throughput
- AIMD-based
 - how to choose AIMD parameters to be TCP friendly

New challenges

- Large bandwidth-delay product networks
 - aka “long-fat” (elephant) networks
 - example by Floyd: “A standard TCP connection with
 - 1500-byte packets,
 - a 100ms round-trip time, and
 - a steady-state throughput of 10Gbps,
 - would require
 - an average congestion window of 83,000 packets and
 - at most one drop (mark) every 5,000,000,000 packets (or equivalently, at most one drop every 1 2/3 hours).
 - This is *not* realistic”

Another example

- Scenarios
 - 10 Gbps point-to-point, dedicated link
 - 1500-byte packets
 - 100 ms round-trip time
 - large enough sender and receiver buffer
- Questions
 - how long does it take to fill the pipe initially?
 - after the first timeout?
 - after the follow-on triple dupack?
 - what is the link utilization?

TCP congestion control

- AI
 - on a new ack
 - $\text{cwnd} = \text{cwnd} + \text{MSS} * \text{MSS} / \text{cwnd}$
 - equivalently, $\text{cwnd} += \text{MSS}$ for every RTT
 - or $\text{cwnd} += \text{MSS} / b$ if acknowledging every b packets
- MD
 - on a loss event
 - $\text{cwnd} = \text{cwnd} / 2$
 - AI follows if Fast Recovery
 - $\text{cwnd} / 2$ RTT to increase from $\text{cwnd} / 2$ to cwnd

Critics on TCP congestion control

- Congestion loss vs transmission error
 - e.g., wireless links
 - approaches: TCP over wireless
 - transport-layer approaches
 - link-layer approaches
 - hybrid approaches
- “(1, 0.5)-AIMD is too conservative/aggressive”
 - Discussion
 - when is (1, 0.5)-AIMD good?
 - when is not?

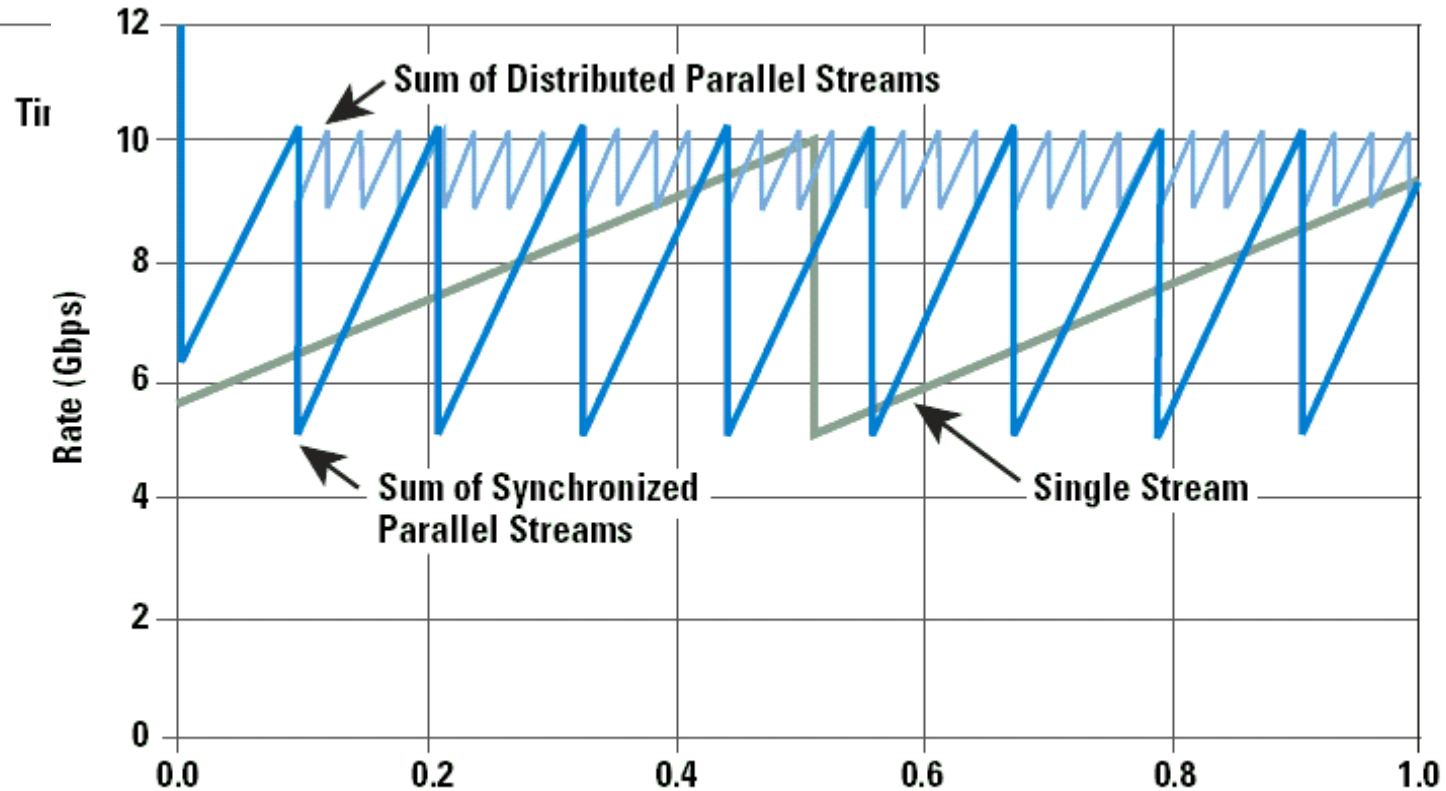
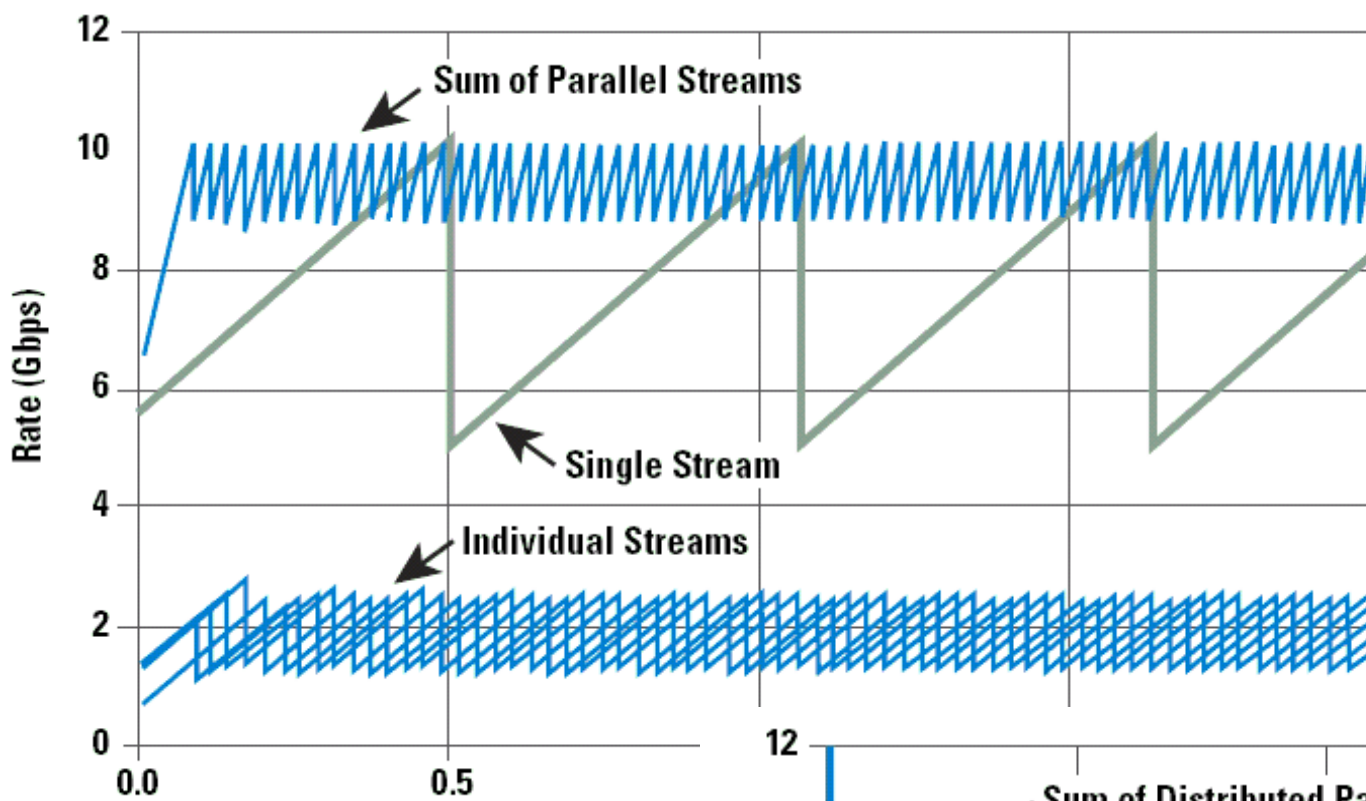
Other issues with “elephant” networks

- Window size
 - TCP: 16-bit window size; byte sequence
 - i.e., 64 KB unacknowledged data at most
 - on high-speed links
 - transmission time \ll propagation time $<$ round-trip time
- Sequence space
 - TCP: 32-bit sequence space; byte sequence
- Approach
 - TCP window scale option
 - left-shift at most 14 bits
 - i.e., 1 GB

Approaches

- Multi-TCP
 - multiple TCP connections
 - between the same pair of endpoints, or
 - from many endpoints to one endpoint (data sink)
 - good: no changes to TCP
 - bad: many TCP connections in one endpoint
 - appropriate data splitting and reassembly
 - ugly: synchronization between connections
- Newer TCP
 - goal: work well in elephant networks
 - also work well with legacy TCP in regular networks

Multi-TCP



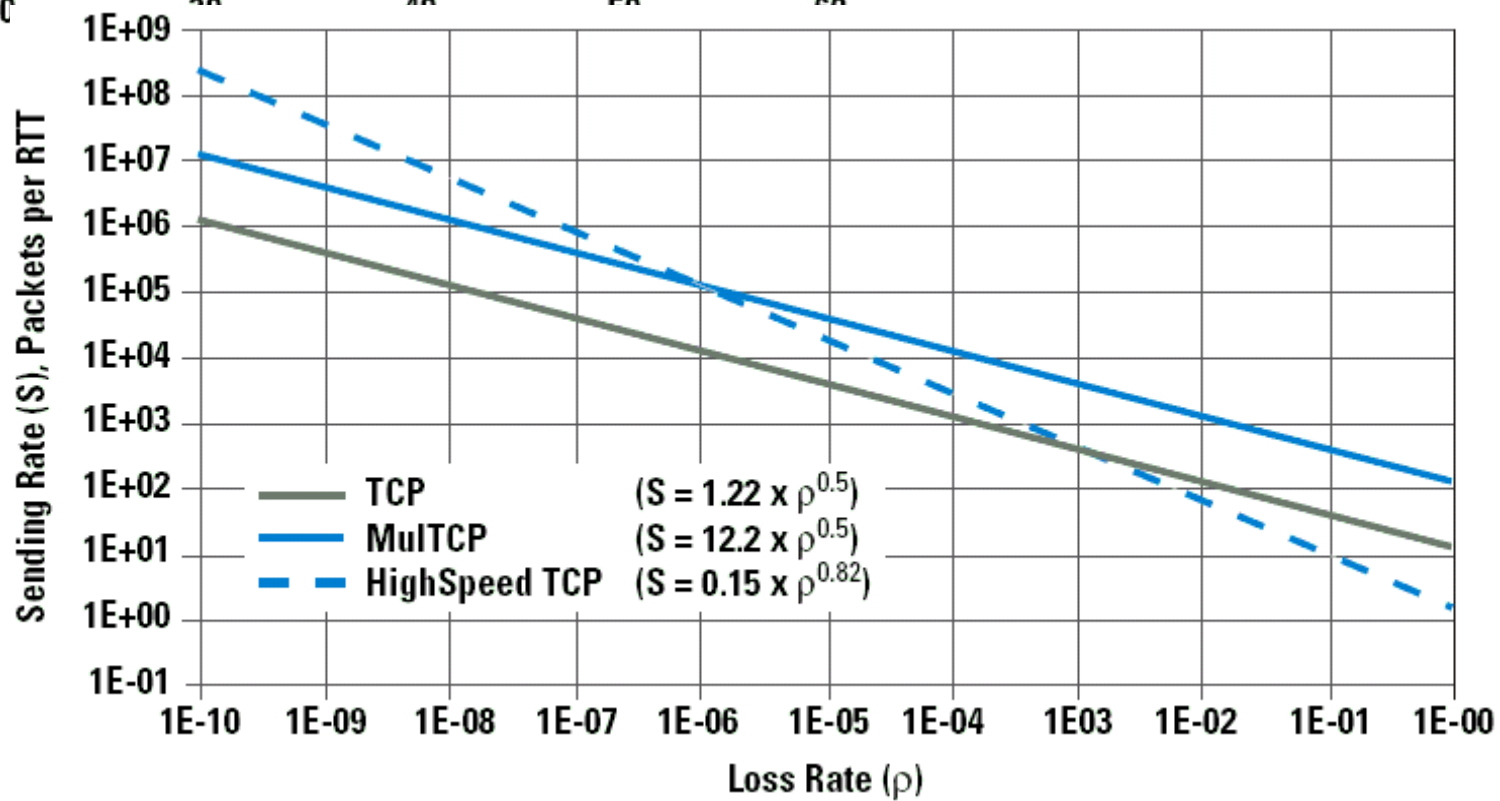
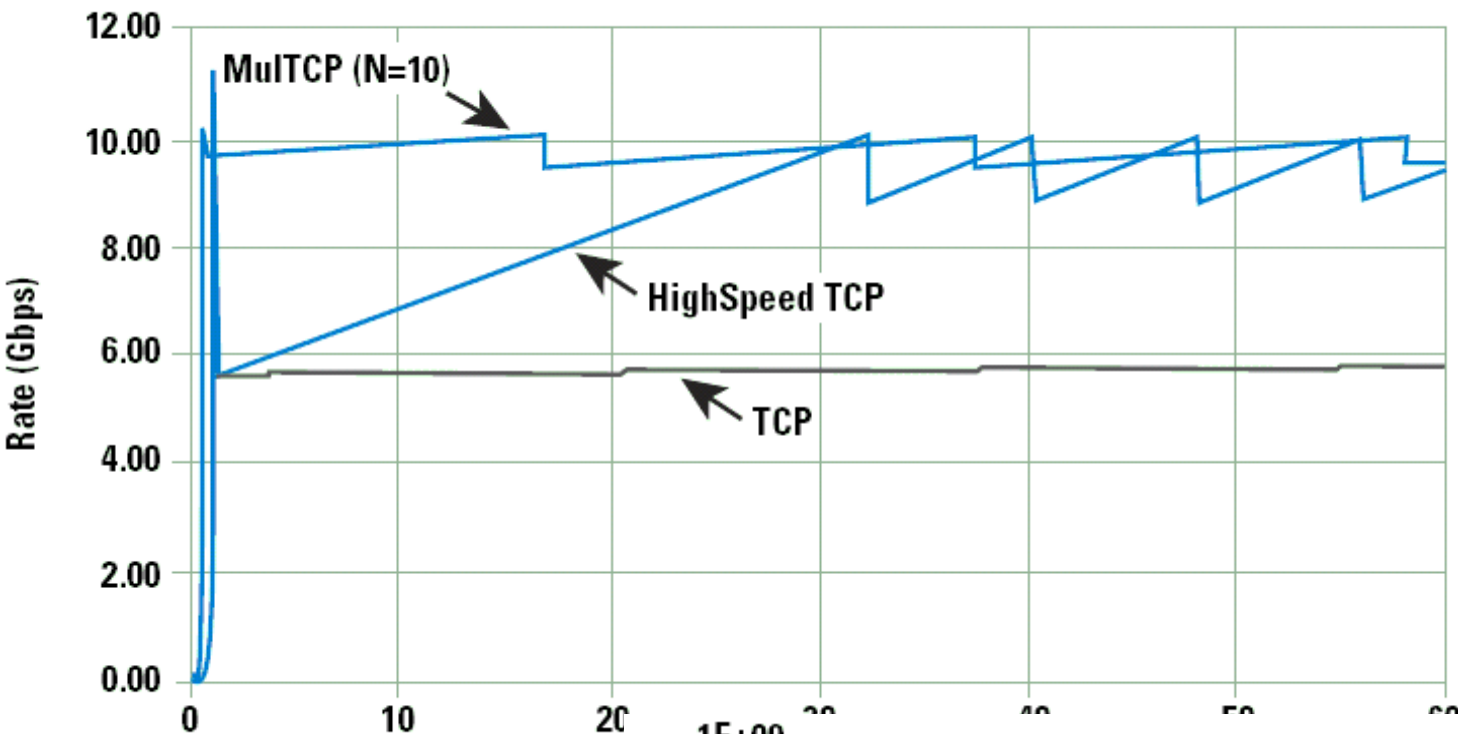
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High-Speed TCP

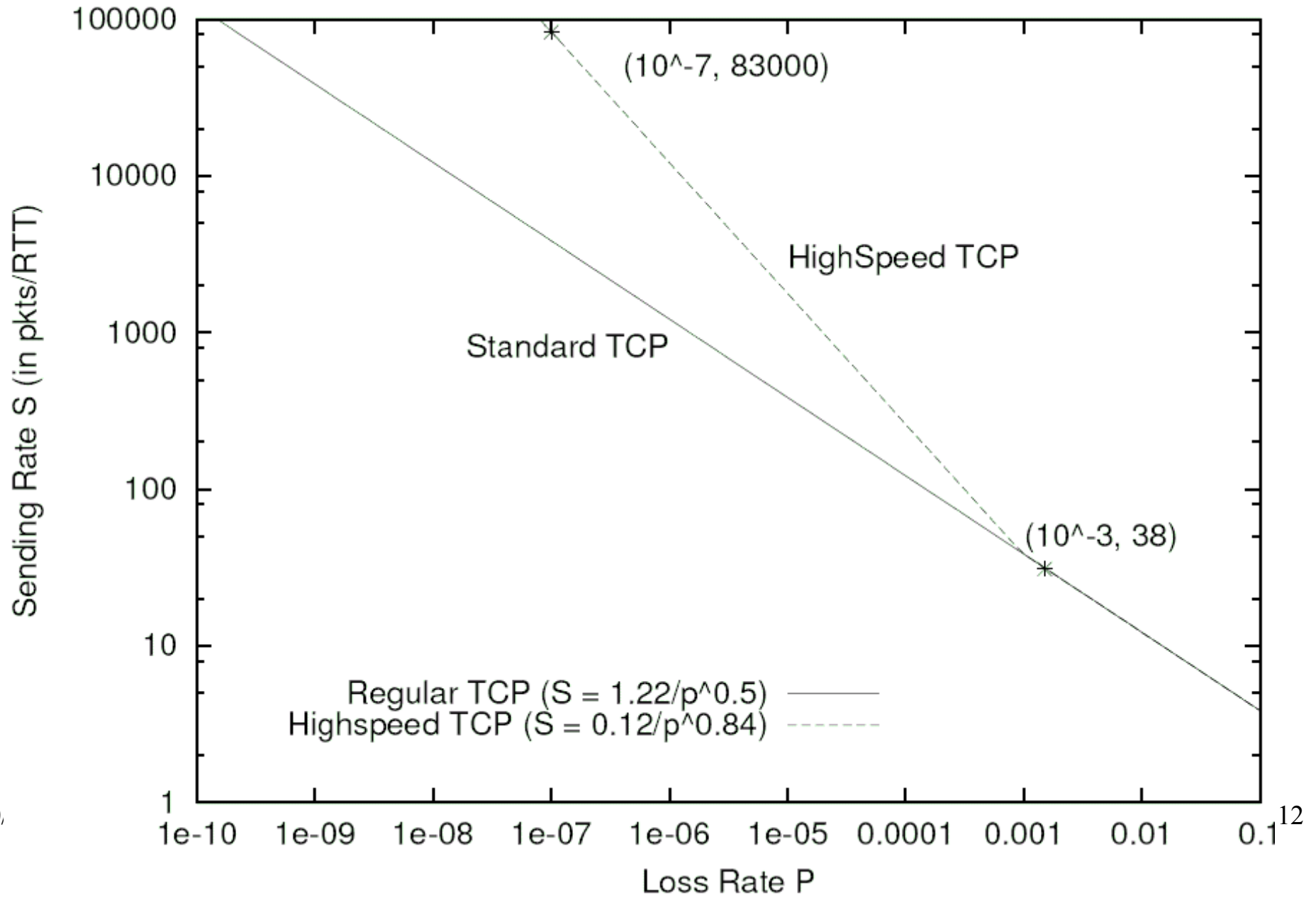
- MuTCP: to emulate N TCP connections
 - AI: increase by $N \cdot \text{MSS}$ per RTT
 - MD: reduce by $1/(2N)$ per loss event
 - not TCP-friendly even in non-elephant networks
- HS-TCP by Floyd
 - AI: increase by $a(\text{cwnd})$ per RTT
 - $a(\text{cwnd})$: a function of cwnd
 - higher cwnd, larger $a(\text{cwnd})$
 - MD: reduce by $b(\text{cwnd})$ per RTT
 - higher cwnd, smaller $b(\text{cwnd})$
 - can maintain TCP-friendly in non-elephant networks

HighSpeed TCP (70ms, 1500 Octet Segments, 10Gbps)

MuTCP, HS-TCP

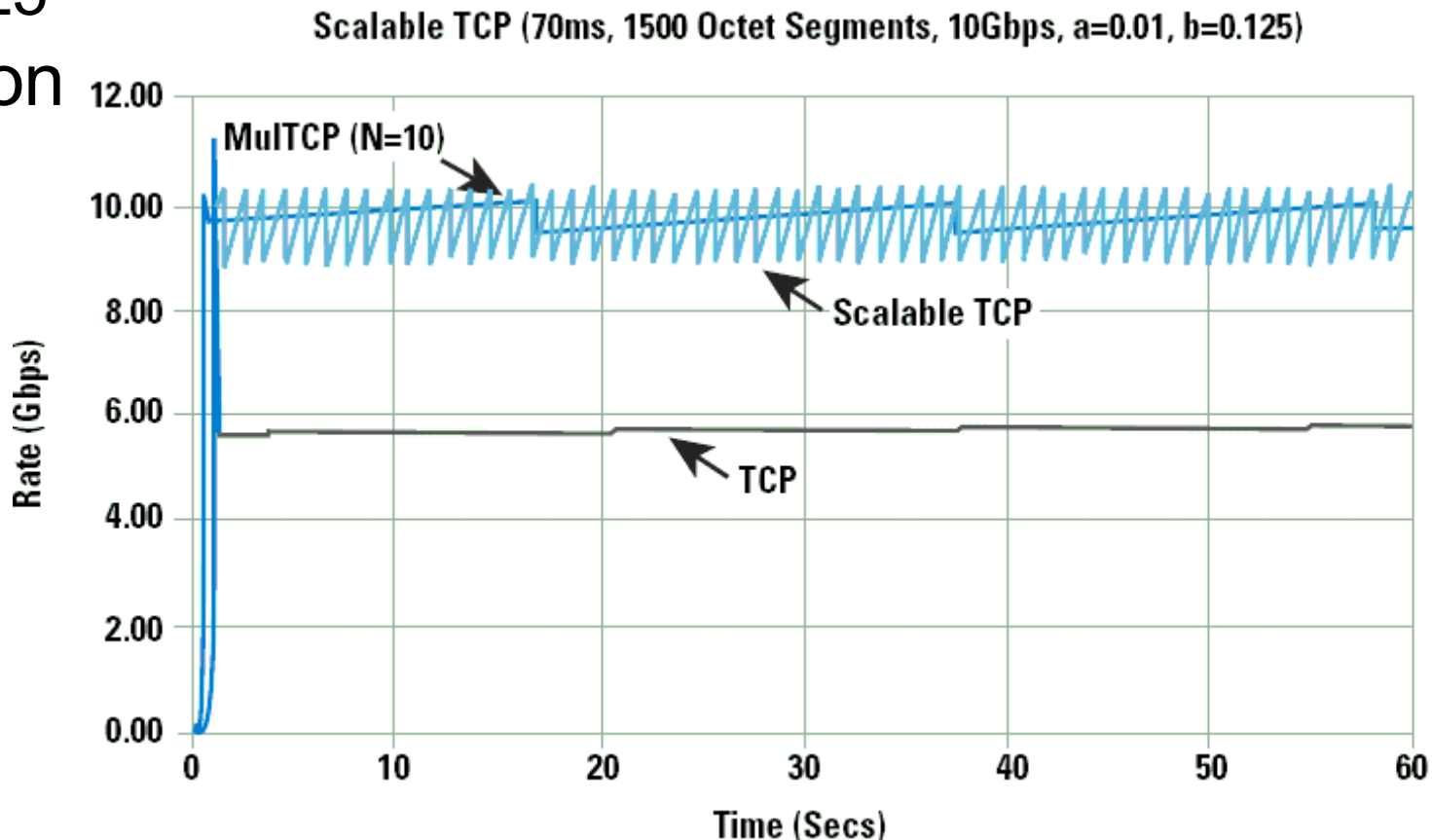


TCP-friendly HS-TCP



Scalable TCP

- S-TCP by Kelly
 - MI: increase by a on each new ack
 - multiplicative increase every RTT; e.g., $a = 0.01$ MSS
 - MD: decrease by b on every loss event
 - e.g., $b = 0.125$
 - more oscillation



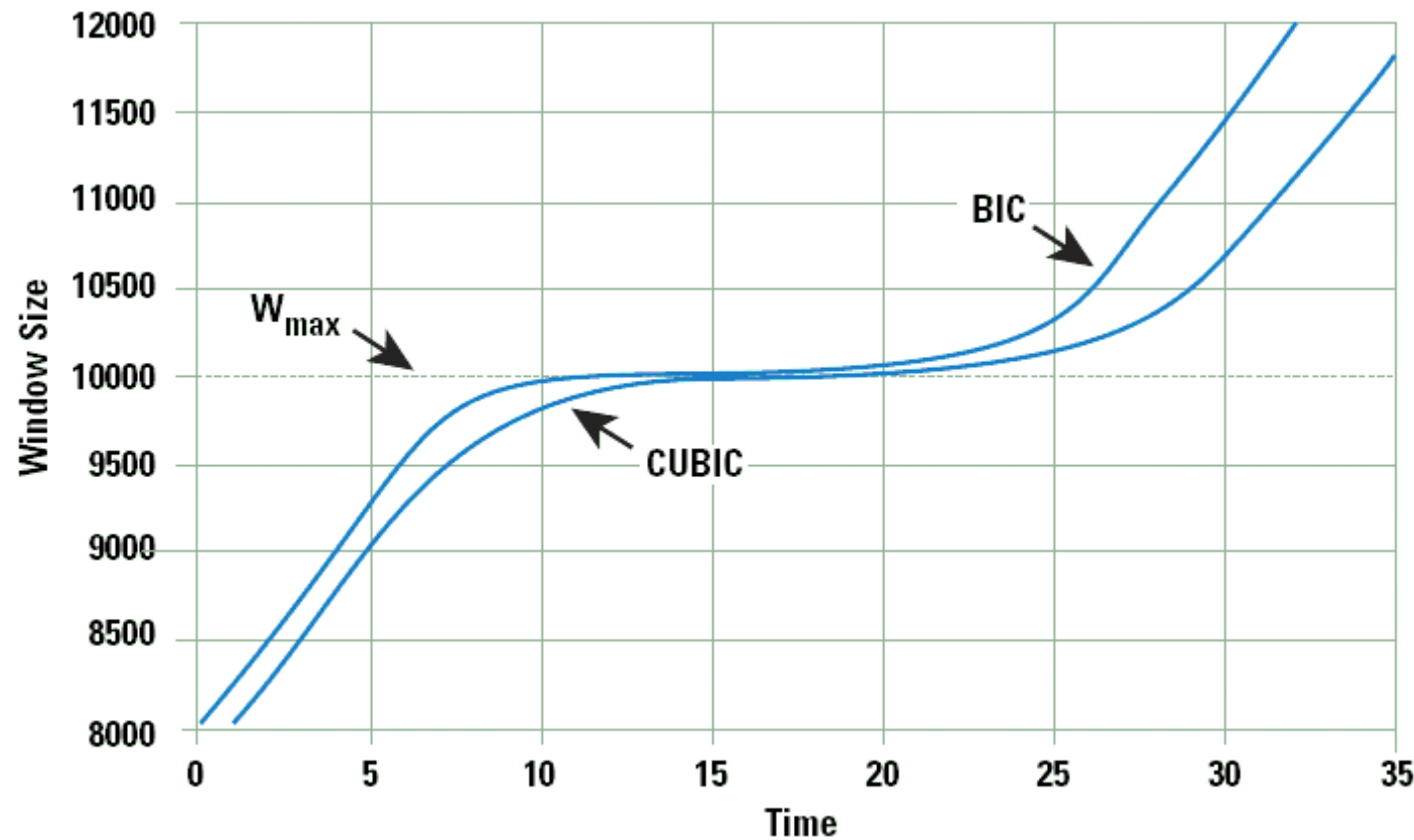
Fast AQM Scalable TCP

- FAST TCP by Low
 - built upon TCP Vegas
 - delay-based congestion control
 - slower than slow-start
 - adjust cwnd every other RTT
 - exit when achievable throughput is lagging behind more a threshold, rather than packet loss
 - multiplicative increase
 - when below equilibrium, approach faster
 - exponential convergence
 - move half-way between the current and target value

BIC and CUBIC

- BIC: binary increase congestion control
 - reduce cwnd on loss event
 - remember cwnd before loss event
 - binary search between current and last cwnd during congestion avoidance

- CUBIC
 - 3rd-order polynomial function
 - better stability



Student Presentation

- Emad Shihab: XCP
 - [KDR02] Dina Katabi, Mark Handley, and Charlie Rohrs. Congestion Control for High Bandwidth-Delay Product Networks. In the proceedings on ACM Sigcomm 2002. [XCP]

Further discussion

- TCP congestion control
 - a long-thriving research thrust
- Network protocols are essentially driven by
 - communication technologies
 - application requirements
 - they often change!

Type	Control Method	Trigger	Response
TCP	AIMD(1,0.5)	ACK response Loss response	$W = W + 1/W$ $W = W - W \times 0.5$
MuITCP	AIMD(N,1/2N)	ACK response Loss response	$W = W + N/W$ $W = W - W \times 1/2N$
HighSpeed TCP	AIMD(a(w), b(w))	ACK response Loss response	$W = W + a(W)/W$ $W = W - W \times b(W)$
Scalable TCP	MIMD(1/100, 1/8)	ACK response Loss response	$W = W + 1/100$ $W = W - W \times 1/8$
FAST	RTT Variation	RTT	$W = W \times (\text{base RTT}/\text{RTT}) + \alpha$

This lecture

- TCP over “long-fat” networks
 - problems and approaches
 - schemes
 - HSTCP, Scalable TCP, FAST
 - XCP
- Explore further
 - Internet Congestion Control Research Group
 - Internet2 Land Speed Record (LSR)
<http://www.internet2.edu/lsr/>
 - Supercomputing Bandwidth Challenge (BWC)

Next lectures

- A new chapter
 - network routing
- Required reading
 - [KZ90] A. Khanna and J. Zinky, "A Revised ARPANET Routing Metric," ACM SIGCOMM '89, pp. 45-56, September 1989.
 - [LMJ97] C. Labovitz, G. R. Malan, and F. Jahanian, "Internet Routing Instability". In Proceedings of ACM SIGCOMM'97, September 1997.
 - [GR00] Lixin Gao and Jennifer Rexford, "Stable Internet Routing Without Global Coordination". In Proceedings of the 2000 ACM SIGMETRICS international conference on Measurement and modeling of computer systems. 2000.