

Analysis of Buffer Size in Core Routers

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Overview

- Problem Description
- Background Information
 - TCP Reno vs. NewReno
 - The Rule-of-Thumb
 - Synchronization
- Experiment Methodology
- Results

Problem Description

- Internet routers buffer packets during times of congestion
- Buffers cause queuing delay
- The fast memory required in core routers is expensive
- Recent studies suggest routers in use today contain buffers larger than necessary

Background Information

TCP Reno vs. NewReno

- TCP is the protocol in use for many applications such as the web, FTP, and email
- There are a number of variants on the TCP protocol
- The studies cited considered only the Reno variant, whereas NewReno is better at recovering from multiple losses

Background Information

TCP Reno vs. NewReno (2)

- TCP is designed to allow multiple connections to share a congested links bandwidth
- Phases
 - Slow Start
 - Congestion Avoidance
 - Fast Retransmit / Fast Recovery

Background Information

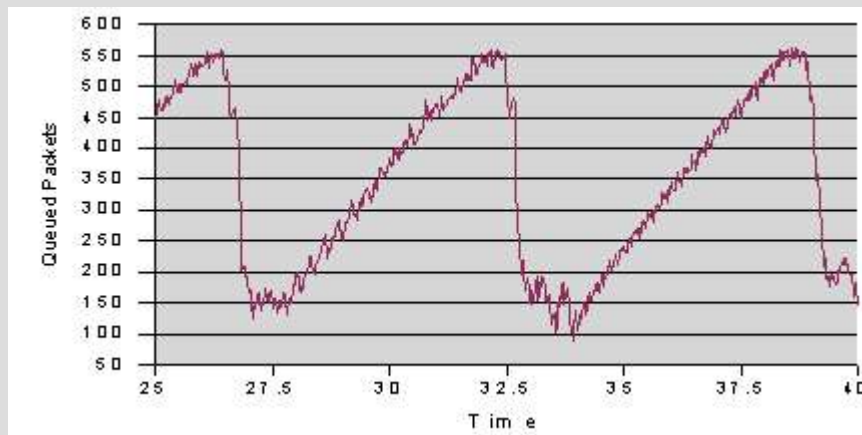
The Rule-of-Thumb

- $B = RTT * C$
- Ensures a congested link does not go idle given a single TCP Reno flow
- Core routers today commonly carry in excess of 10 000 TCP flows

Background Information

Synchronization

- If the buffer size is large:
 - All flows tend to lose packets at the same time
 - So all flows will reduce their sending rate at the same time
- Therefore the sum of all the flows tends to act as a single amplified flow

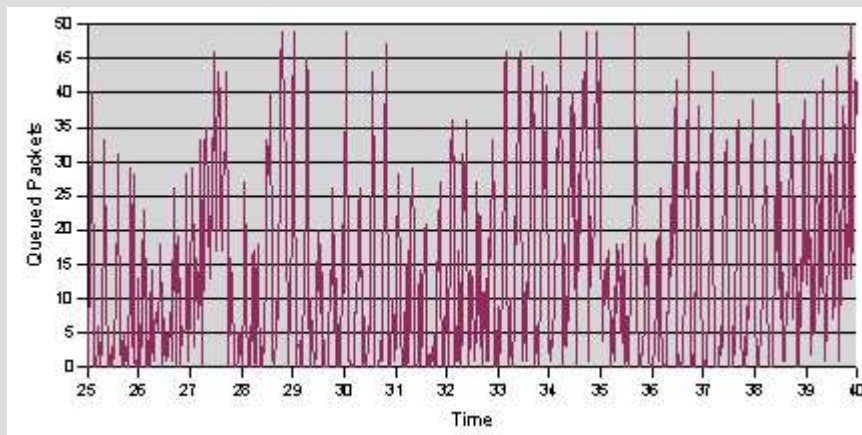


Example: 30 flows through a router with a buffer size of 562 packets

Background Information

Synchronization (2)

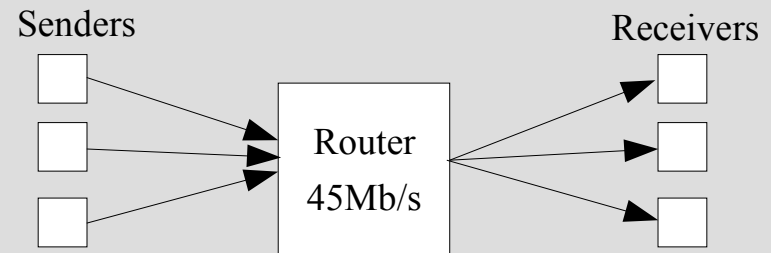
- If the buffer size is small:
 - Flows do not all lose packets at the same time
 - There is no need to buffer packets to keep the link busy as there is no common sending rate reduction
- The queue length will fluctuate rapidly



Example: 30 flows through a router with a buffer size of 51 packets

Experiment Methodology

- Simulation tool: ns-2
- Single congested link
- 45Mb/s data rate on the link
- Constant RTT of 0.1s
- Varied number of flows from 30 to 120



Experiment Methodology (2)

- The paper “Sizing Router Buffers” suggests a buffer size $B = \text{RTT} * C / \text{sqrt}(n)$
- Our buffer size was determined as multiples of the above rule ($1/2x$, $1x$, $2x$, $3x$)
- The resulting utilization was measured for both Reno and NewReno flows

The Results

- The rule $B=RTT \cdot C / \sqrt{n}$ resulted in high utilization of the congested link
- Additional buffer reductions are possible assuming TCP NewReno flows

TCP Flows	RTT*C/sqrt(n)	Buffer Packets	Utilization Reno	Utilization NewReno
30	0.5x	51	82.1%	90.0%
30	1x	103	89.6%	92.3%
30	2x	205	94.7%	96.6%
30	3x	308	97.7%	98.8%
60	0.5x	36	80.5%	86.2%
60	1x	73	91.9%	94.8%
60	2x	145	92.1%	95.7%
60	3x	218	96.8%	98.5%
90	0.5x	30	82.0%	86.8%
90	1x	59	95.0%	96.9%
90	2x	119	92.0%	96.2%
90	3x	178	90.5%	98.0%
120	0.5x	26	81.4%	88.3%
120	1x	51	95.5%	97.3%
120	2x	103	91.5%	98.2%
120	3x	154	87.6%	97.9%

Thank You

Questions?