

Improvements to and Performance of Conservative Mode for Single Choke Fairness Mechanism Specified in Draft 2.0

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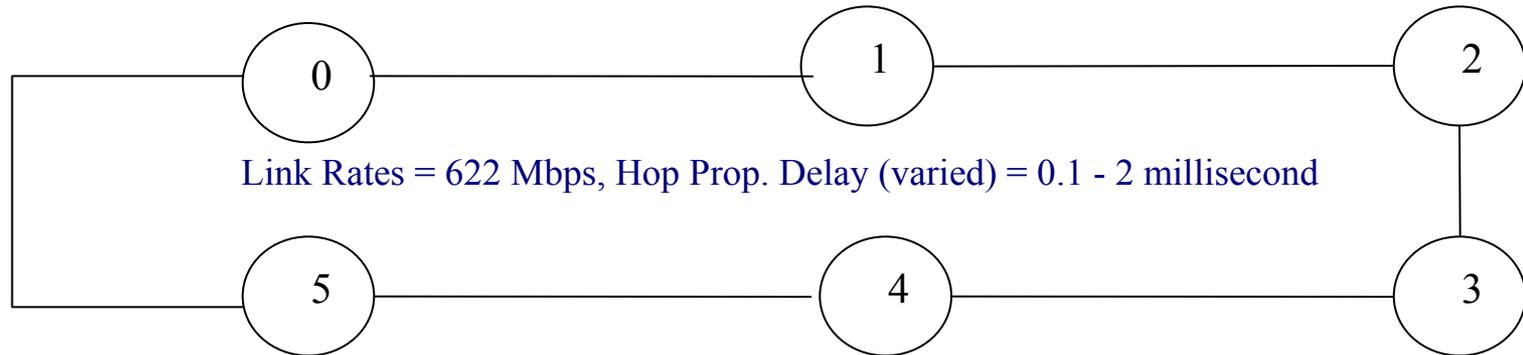
- ❑ At the November IEEE 802.17 meeting, we presented our observations on the single choke aggressive mode for TCP.
- ❑ In certain scenarios, performance showed short term unfairness and oscillations
- ❑ Primary reason for oscillatory behavior was a downstream node's low add rate
 - Upstream node limited to unfair and unreasonably low rate over short term
- ❑ We speculated that computing “fair rate” based on the stations sharing bottleneck link, and communicating this fair rate to upstream nodes would:
 - Allow upstream node's rate to be determined by the actual link bandwidth available
 - Avoid starvation (complete shut-down) of sources for brief periods of time
 - Mitigate oscillations and potentially improve link utilization
- ❑ Conservative mode appeared to have these properties
- ❑ We focused on the performance of dual transit buffer, conservative scheme
 - Performance of existing specification, with simplifications - when STQ present
 - Determined several small, but critical, improvements were needed
 - Significant improvement in performance observed with our suggested modifications

Our high level understanding of existing single choke conservative scheme

- ❑ Differences between aggressive and conservative schemes
 - Primarily in congestion detection and local fair rate calculation
- ❑ When a span is “congested”, backpressure mechanism using Fairness Control Messages (FCM) is triggered
 - STQ buffer occupancy rises above “low threshold” *OR* (filtered) Transmit rate above “low threshold” \Rightarrow “Congested”
- ❑ Fair rate calculation is based on active stations in the “congestion domain”
- ❑ Hysteresis in STQ length used to increase/decrease local Fair Rate
 - Congested node modifies local Fair Rate **every round-trip time**:
 - ❖ Increase when STQ is below “low threshold”, based on ramp coefficient
 - ❖ Decrease when STQ is above “high threshold”, based on ramp coefficient
 - Maintain local Fair rate as is when STQ is in-between low and high thresholds
- ❑ While congested, the node sends local Fair Rate in FCM every “advt. interval”
- ❑ When congestion clears, upstream nodes allowed to ramp up to “unreserved rate”
 - STQ buffer sufficient to receive packets in transit, to accommodate feedback delay

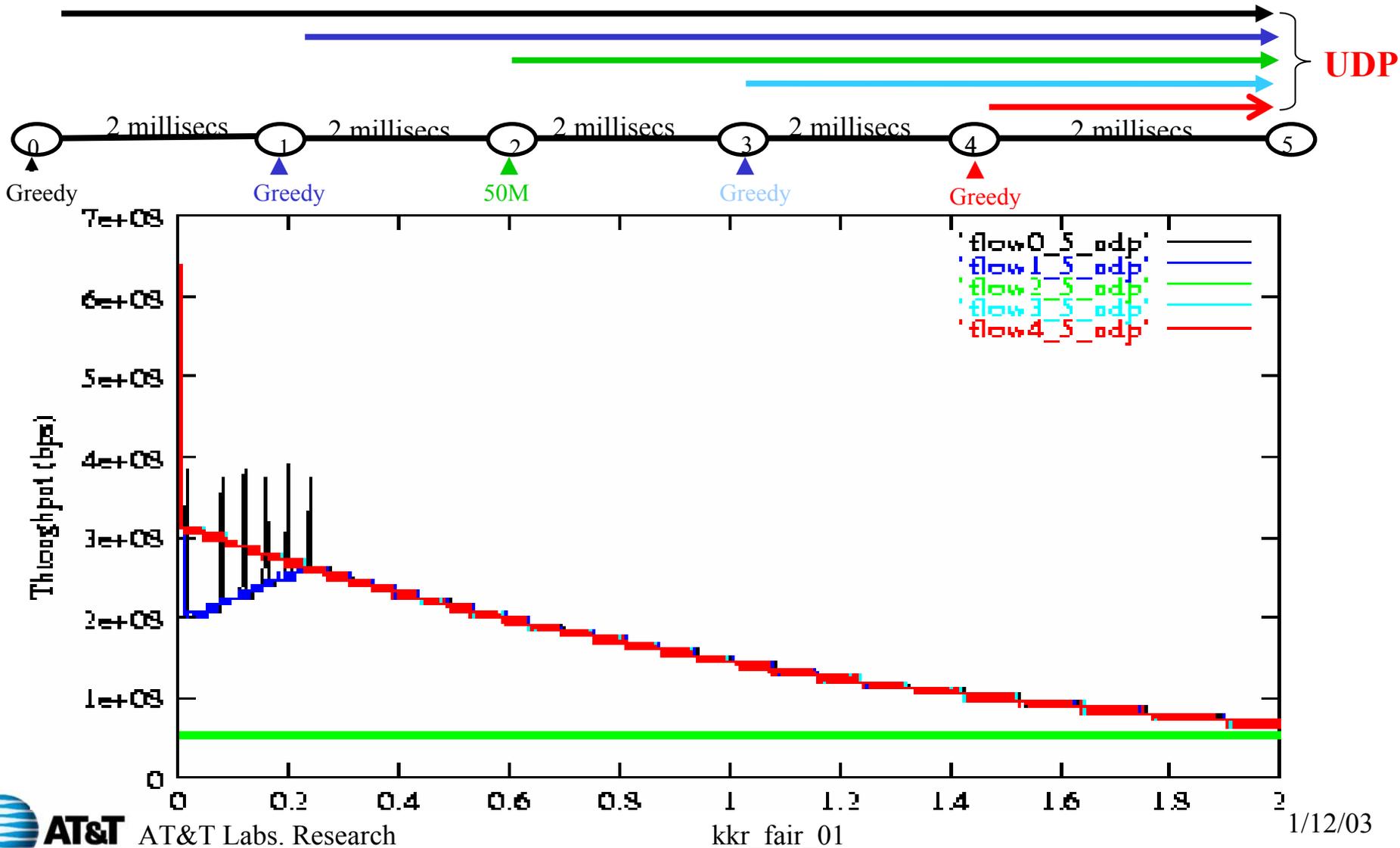
- ❑ Achieve high utilization on the ring – fairness eligible traffic uses available bandwidth opportunistically
 - Allow sources to start up at high rate
- ❑ Converge to fair value as quickly as possible
 - Should achieve fairness on per-source basis with single bottleneck
 - Preferable if unused bandwidth at bottleneck is shared by other sources
- ❑ Operate over a wide dynamic range
 - Even nodes with small STQ buffer still function reasonably over a large ring
- ❑ Minimize oscillations in throughput as much as possible
 - Reflected by the “allowed rate” at the RPR level
 - Reflected at application level by throughput and other specific parameters (e.g., window size for TCP)
- ❑ Avoid even short-term starvation of individual nodes caused by congestion control actions

Experimental Framework

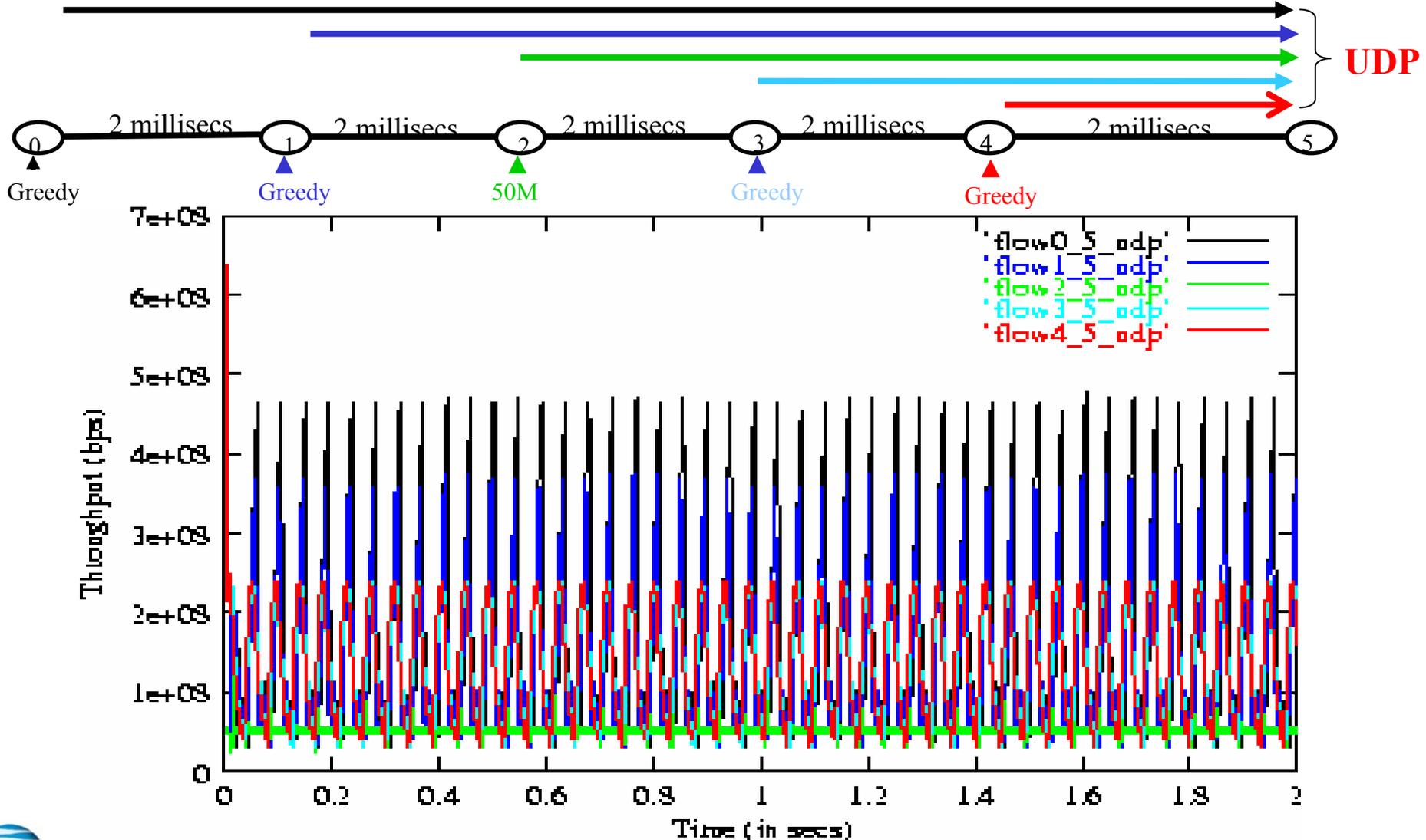


- ❑ Simulations based on NS-2 simulator (initial version from Rice University)
- ❑ Most of simulation results presented here based on a 6 node ring
 - Link rates = 622 Mbps; prop. delay = 0.1-2 millisecs; AgingInterval=200 μsecs; Advt. Interval = 100 μsecs.
 - STQ = 256 Kbytes; Client buffer = 1000 packets (pkt. size = 1040 bytes)
 - Single congestion domain, with one link being the bottleneck
- ❑ Experiments:
 - Several experiments with UDP (constant rate) flows
 - Steady (greedy) TCP flows (FTP); max. window size =64
 - Mixture of TCP and UDP flows
 - Start/stop (short-lived) flows: shows scheme's dynamics and responsiveness

- Simulation: constant rate (UDP) sources; Default parameters; fixed “RTT” = 20msecs.
- Did not include rate thresholds



- Same experiment as previous slide, with 5 UDP flows, 2 millisecc. Per-hop propagation delay





Outline of Proposed Improvements



- ❑ We have maintained the basic framework of conservative scheme as defined in Draft 2.0
- ❑ Our efforts to improve the performance of the existing Draft 2.0 conservative scheme fall into two broad areas
 - Improved estimate of LocalFairRate
 - ❖ In existing specification, goal for initial value of LocalFairRate is equal share: “unreserved rate/active stations”
 - Means to estimate Round Trip Time within a Congestion Domain
 - ❖ Existing specification uses “RTTWorthofIntervalsHavePassed”, but does not specify how this is obtained, and what RTT refers to.
- ❑ We have focused on the Dual Transit buffer scheme
 - Simplifications involve looking only at STQ buffer size
 - ❖ Do not have targets for utilizations, as reflected by Rate Thresholds
- ❑ No new or additional variables are introduced
- ❑ Also improved the overall weighted fair allocation mechanism
 - Applicable to both aggressive and conservative scheme





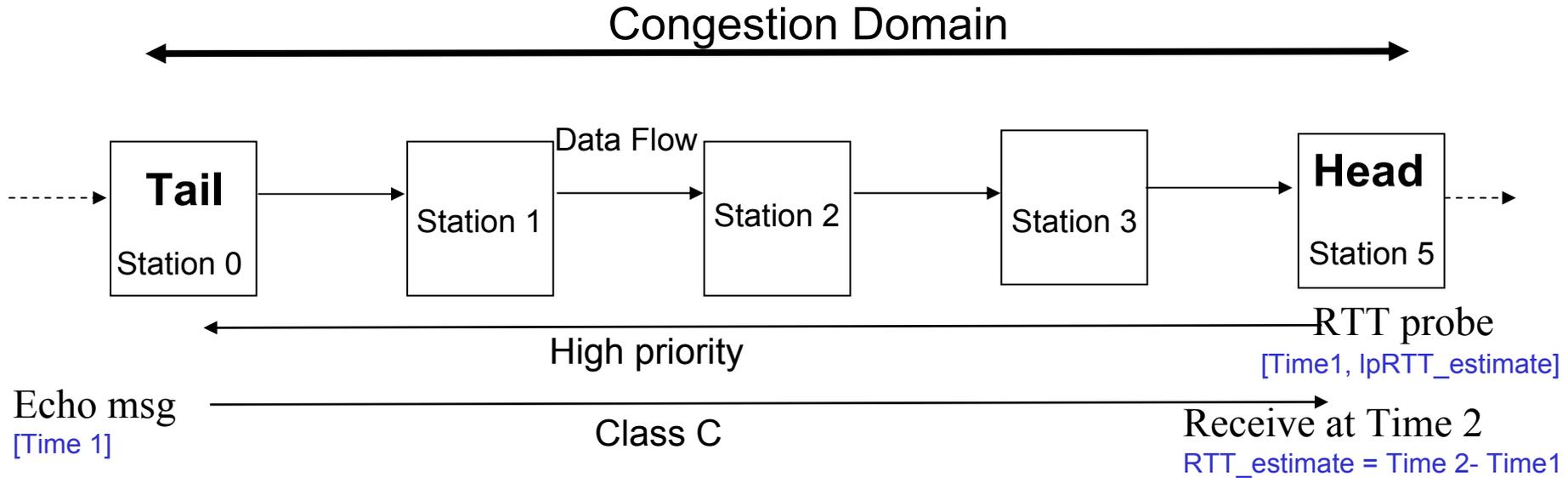
Enhanced LocalFairRate Calculation



- ❑ Scheme was designed to work with a reasonable initial value
 - We have improved the initial value of LocalFairRate slightly to factor in the case when the demand of local station is less than “fair share”
 - ❖ Helps in reducing oscillations and convergence time
- ❑ LocalFairRate is calculated every round trip time, as spec'd before
- ❑ During the period a node is congested, we introduce two simple modifications to LocalFairRate computation to reduce oscillations
 - If $STQDepth > STQHighThreshold$, LocalFairRate reduced by a factor
 - ❖ We add a lower bound to LocalFairRate based on what fair share should be
 - If $STQDepth < STQLowThreshold$, LocalFairRate is increased by a factor
 - ❖ We improve the value LocalFairRate is increased by, knowing the amount of traffic added by the local station
- ❑ New sources starting up at high rate can starve downstream stations whose STQ buffer is full due to rapid onset of congestion
 - If this happens, station recomputes LocalFairRate when local congested station has a lower add rate than the fair share, and advertises this updated rate



- ❑ Conservative scheme specified in Draft 2.0 updates LocalFairRate (increase/decrease) every RTTWorthofIntervalsHavePassed
- ❑ Having all the nodes update their localFairRate at the same fundamental frequency of the control loop (RTT) gives a stable and properly damped control
 - When local state is congested or uncongested
 - Recompute rate only after previous change in the rate has taken effect.
- Control loop RTT includes:
 - ❖ the instant at which LocalFairRate is advertised by a congested node in an FCM
 - ❖ the FCM is received by all the nodes in the congestion domain
 - ❖ these nodes adjust their allowedRateCongested
 - ❖ congested node observes results of this change in its received FE traffic
- ❑ RTT is estimated for the congestion domain:
 - “Head” (next to congested span) to “Tail” (last upstream node of domain)
 - Estimated by Fairness Control Unit.
 - Measured RTT is smoothed by a low pass filter



- ❑ Head station sends a message upstream with $\text{Time1} = \text{current time}$, and low pass filtered RTT_estimate (lpRTT_estimate)
 - A high priority message to emulate transport of FCM
- ❑ Upstream stations in congestion domain copy lpRTT_estimate
- ❑ Tail station echoes back timestamp, Time1 in a message to Head
 - A low priority message (Class C) to emulate transport of FE traffic
- ❑ Head station computes RTT_estimate and performs low pass filter

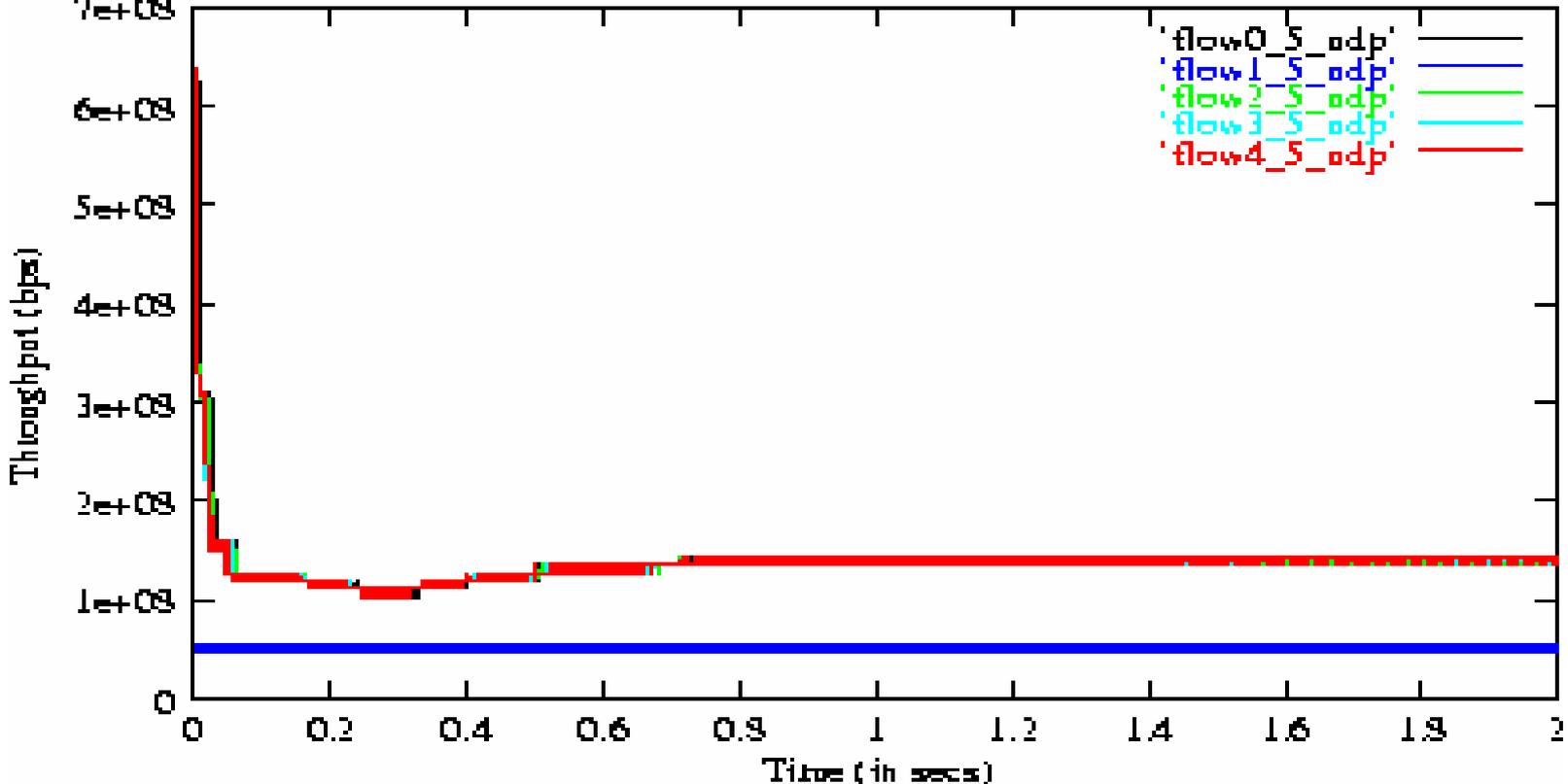
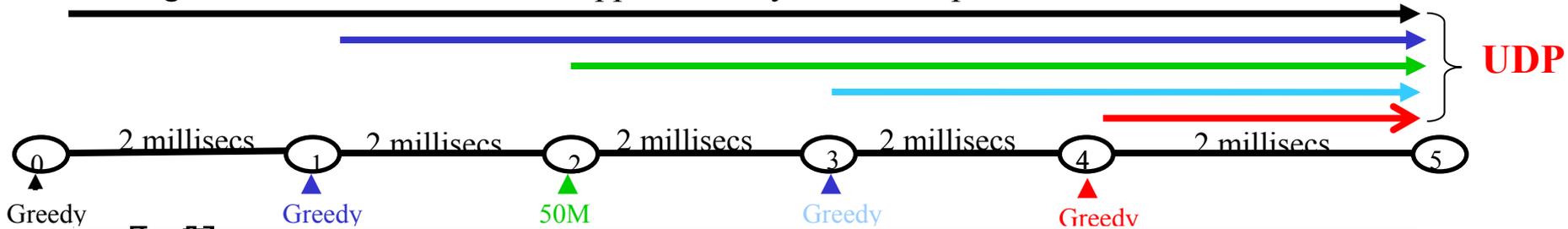
- ❑ When calculating the LocalFairRate, the estimate of fair share is:
 - $\text{unreserved_rate}/\text{activeWeights}$
- ❑ Allows for weighted fair allocation, rather than equal share
 - may have been oversight in specification
- ❑ Since lpAddRate of local station is known, further improve estimate
 - $(\text{unreserved_rate} - \text{lpAddRate})/(\text{activeWeights} - \text{WEIGHT})$
- ❑ In Table 9.5, we needed to make sure that the Fairness Control Message was propagated appropriately
 - Simple change (in Row 1) to cover case: $\text{normLocalFairRate}=\text{rcvdFairRate}$
 - Ensure that FCM was propagated all the way to the “Tail” node of the congestion domain
- ❑ These changes removed some of the problems we had been encountering with weighted fair allocation for both aggressive and conservative modes



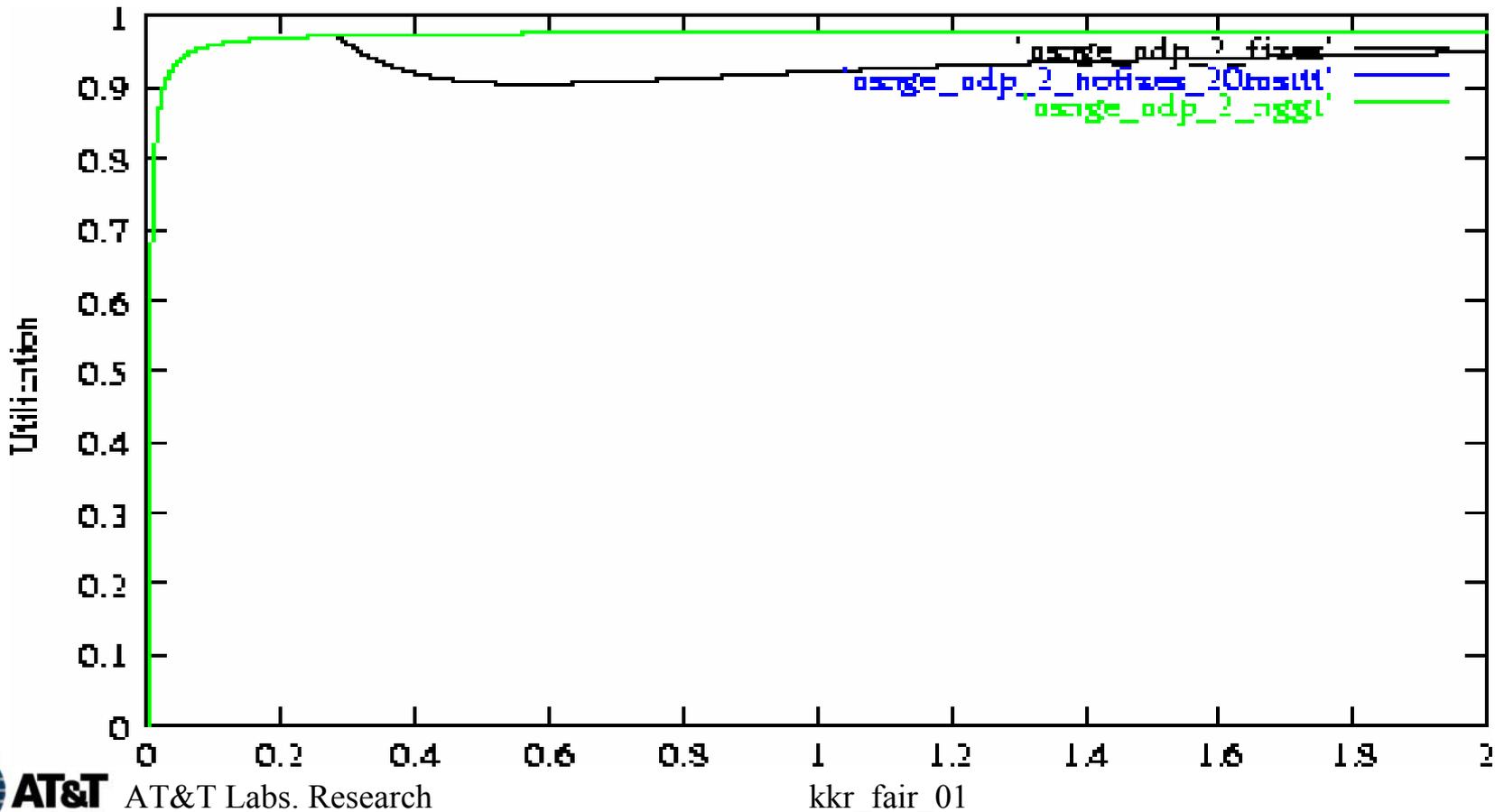
Enhanced Conservative Scheme Performance



- Simulation: constant rate (UDP) sources; Default thresholds; agingInterval = 200μsecs.)
- Convergence time to “fair share” is approximately 4 round trips



- Oscillations in individual station's throughput considerably reduced with new scheme
 - Initially, converges to “equal share” (120 Mbps). Algorithm actually accounts for lower demand from station 2, and gradually approaches max-min fair share = $(622-50)/4 = 144$ Mbps
 - ❖ Max-min fairness (i.e., per-flow fairness) will not be achieved in all cases though
- Bottleneck link utilization for enhanced conservative scheme is below the aggressive and original conservative schemes, when averaged over time

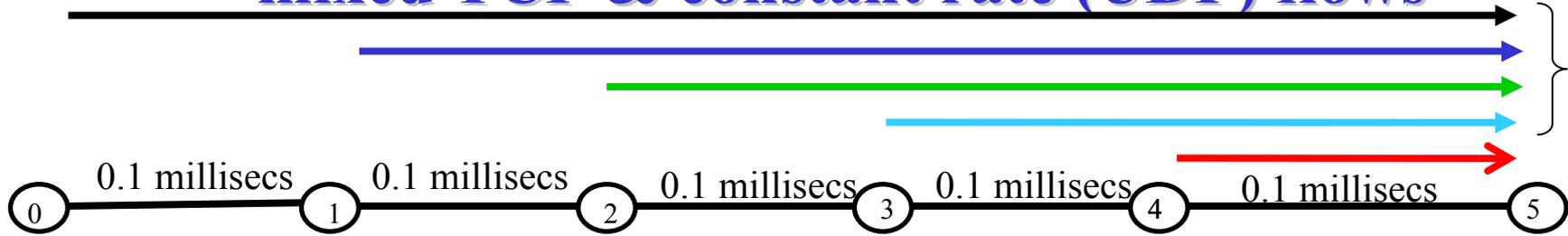




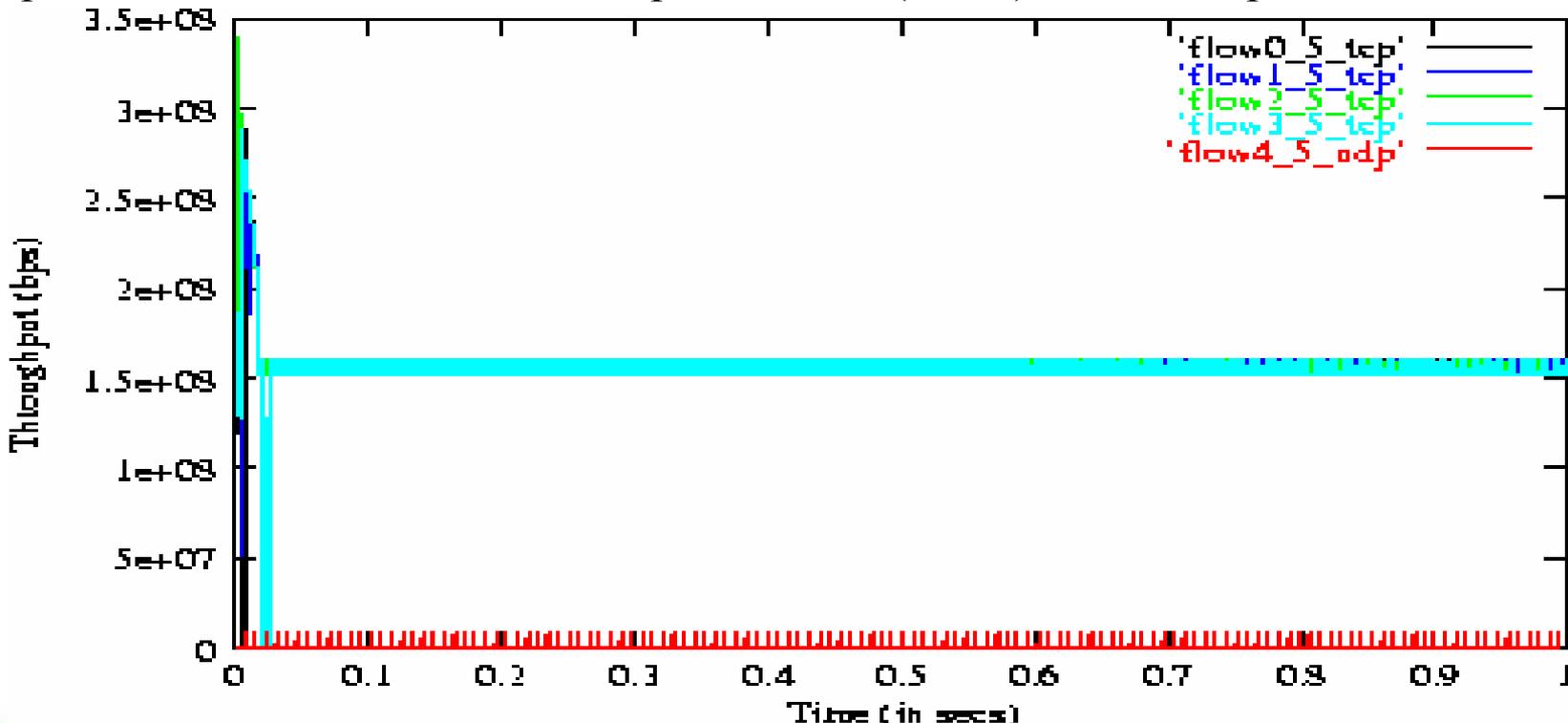
Performance of Enhanced Conservative with mixed TCP & constant rate (UDP) flows

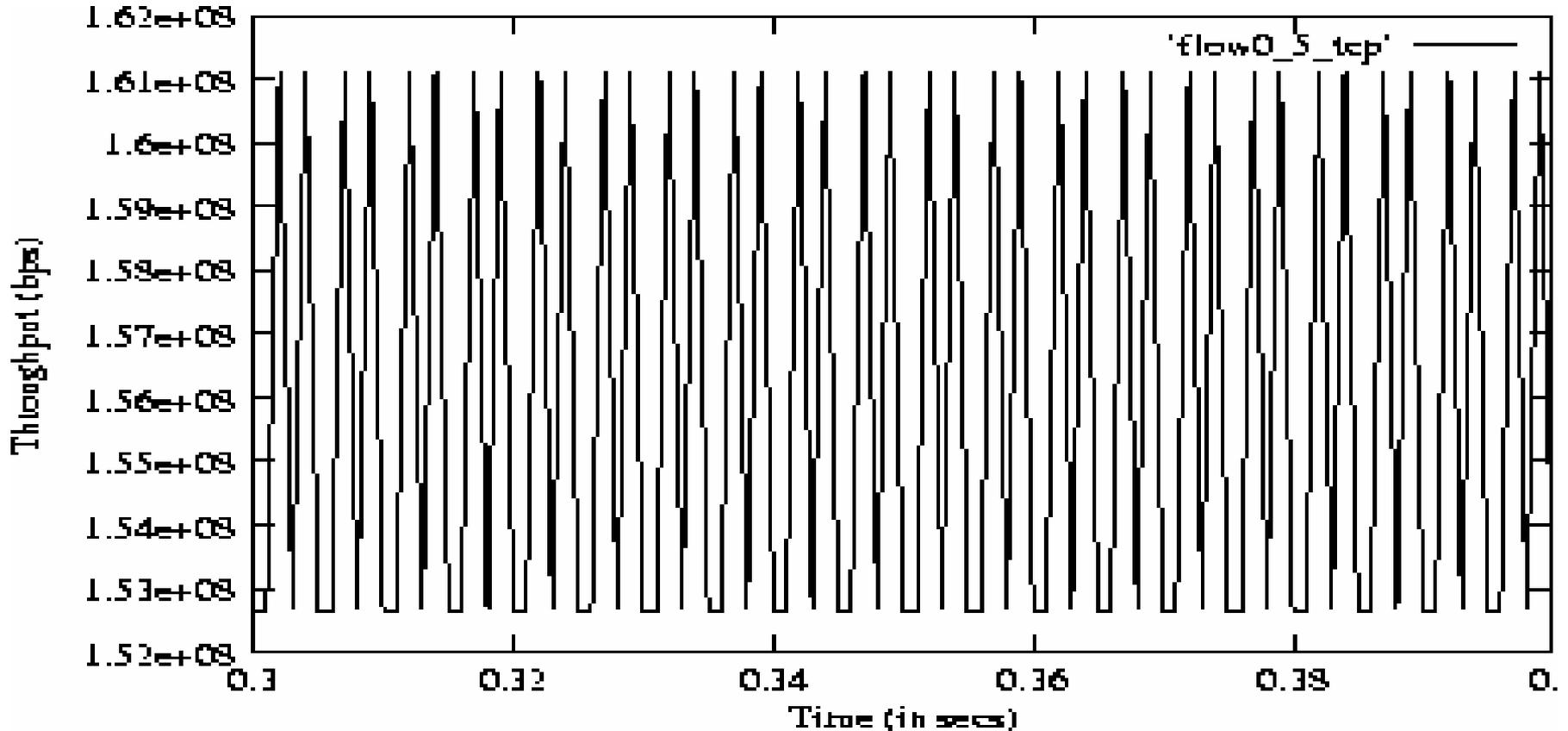


10 Greedy
TCPs each
1Mbps
UDP



- UDP flow is constant rate (=1Mbps) on last hop
- 10 greedy TCP flows from each RPR node; TCP co-resident w/MAC client
- Span 4 → 5 is bottleneck; Fair rate per source = $(622-1)/4 \approx 155$ Mbps



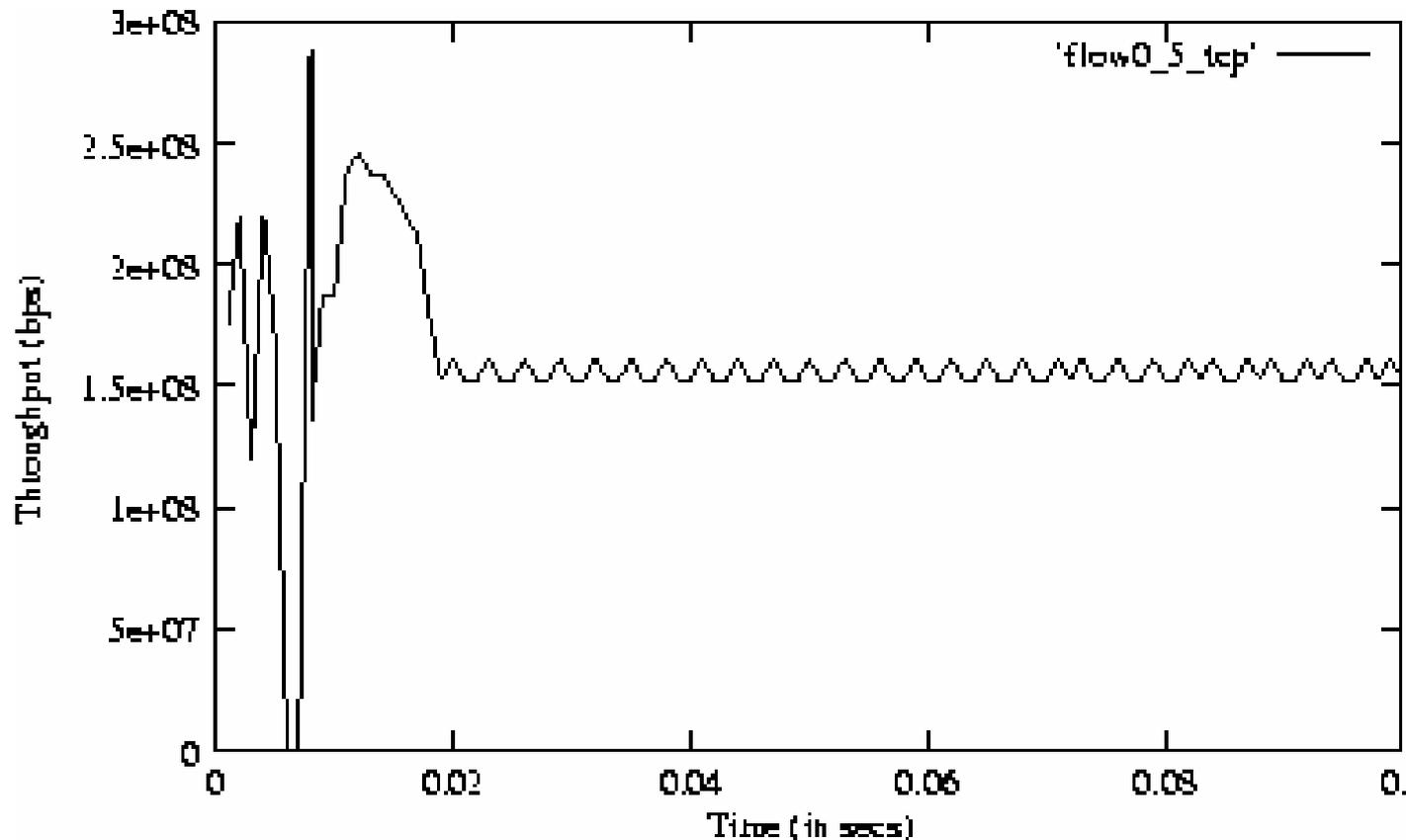


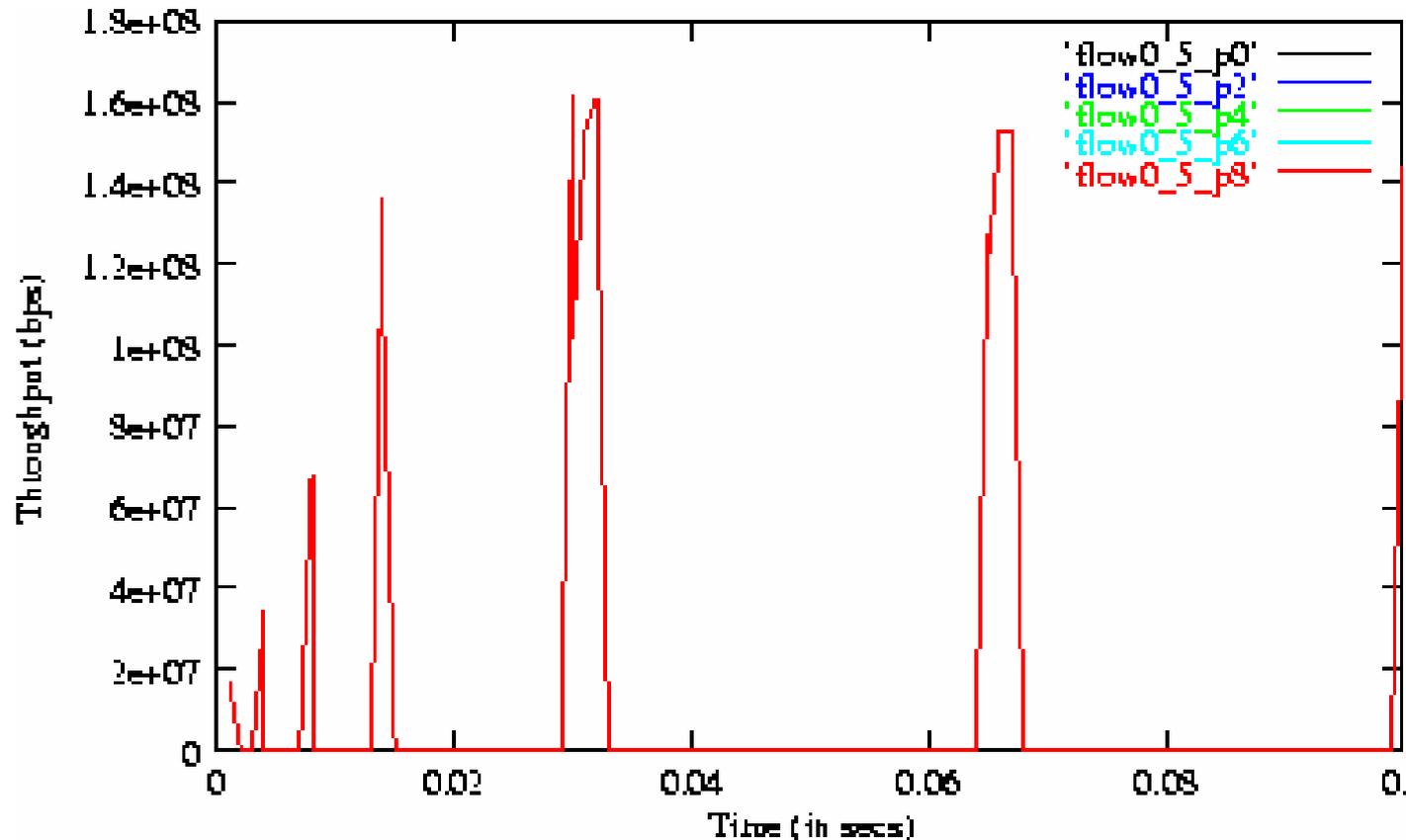
Short term behavior of aggregate of 10 TCPs from node 0

- Individual node throughput oscillates over a tight range

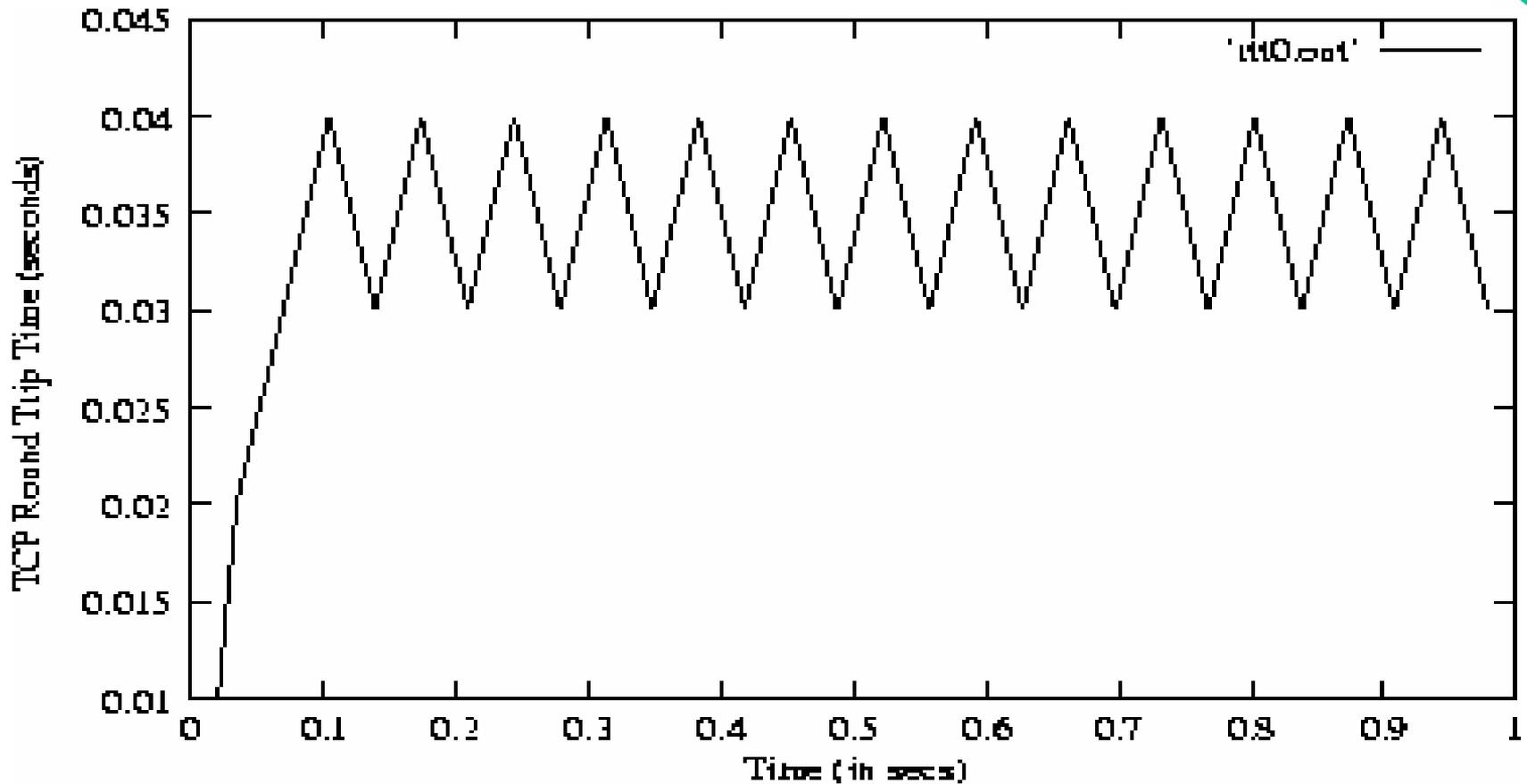
Initial short-term Throughput for one node

- Initial startup behavior to show convergence times
- Initial convergence to the fair, stable rate occurs within 20 milliseconds
 - Primarily driven by TCPs being in slow start and waiting for acks. to grow the window

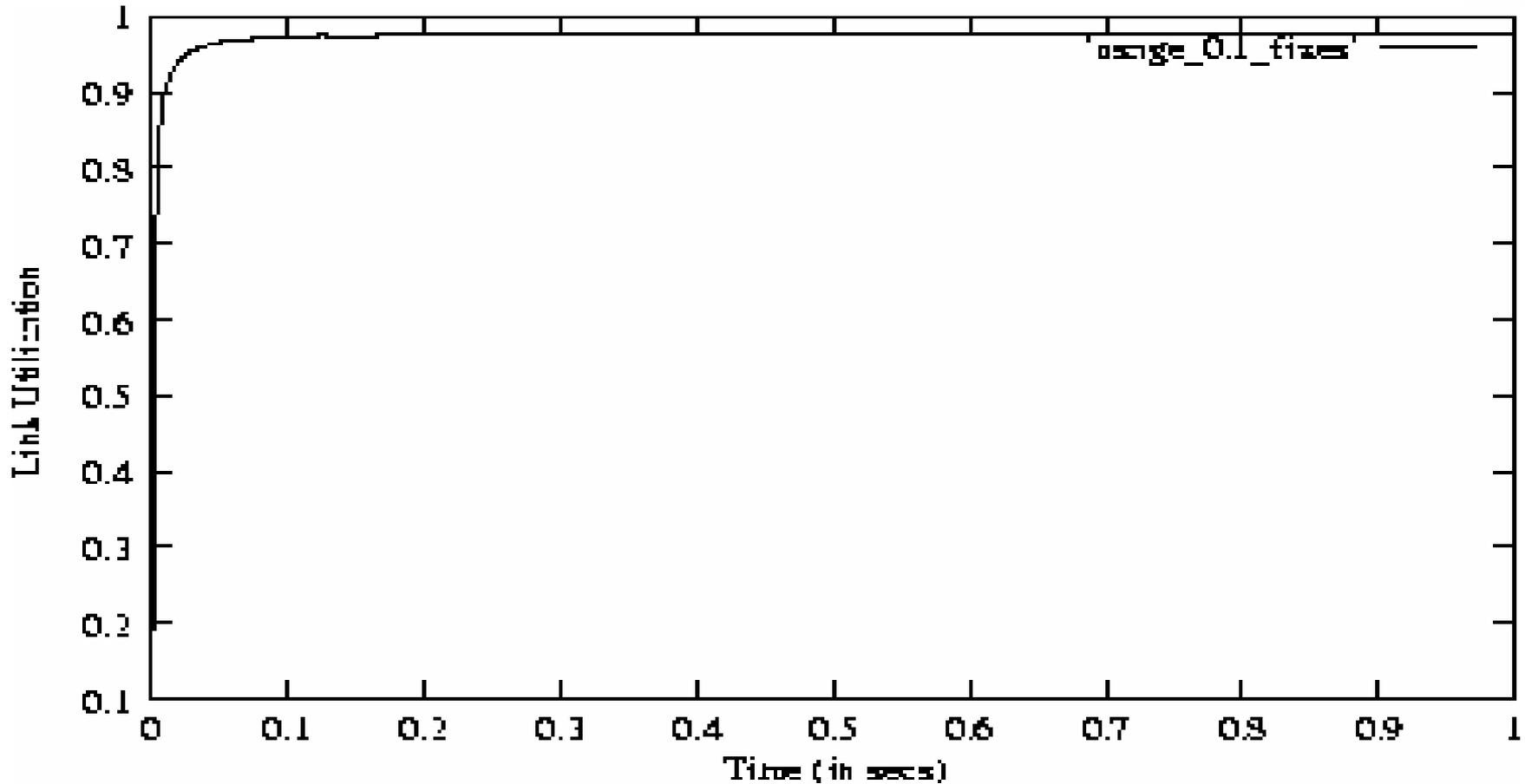




- Each individual TCP generates packets when acks are received
 - Overlaps with the other TCPs at source in generating packets (buffered in client)
 - Startup behavior shows TCP slow start action, exponentially increasing window

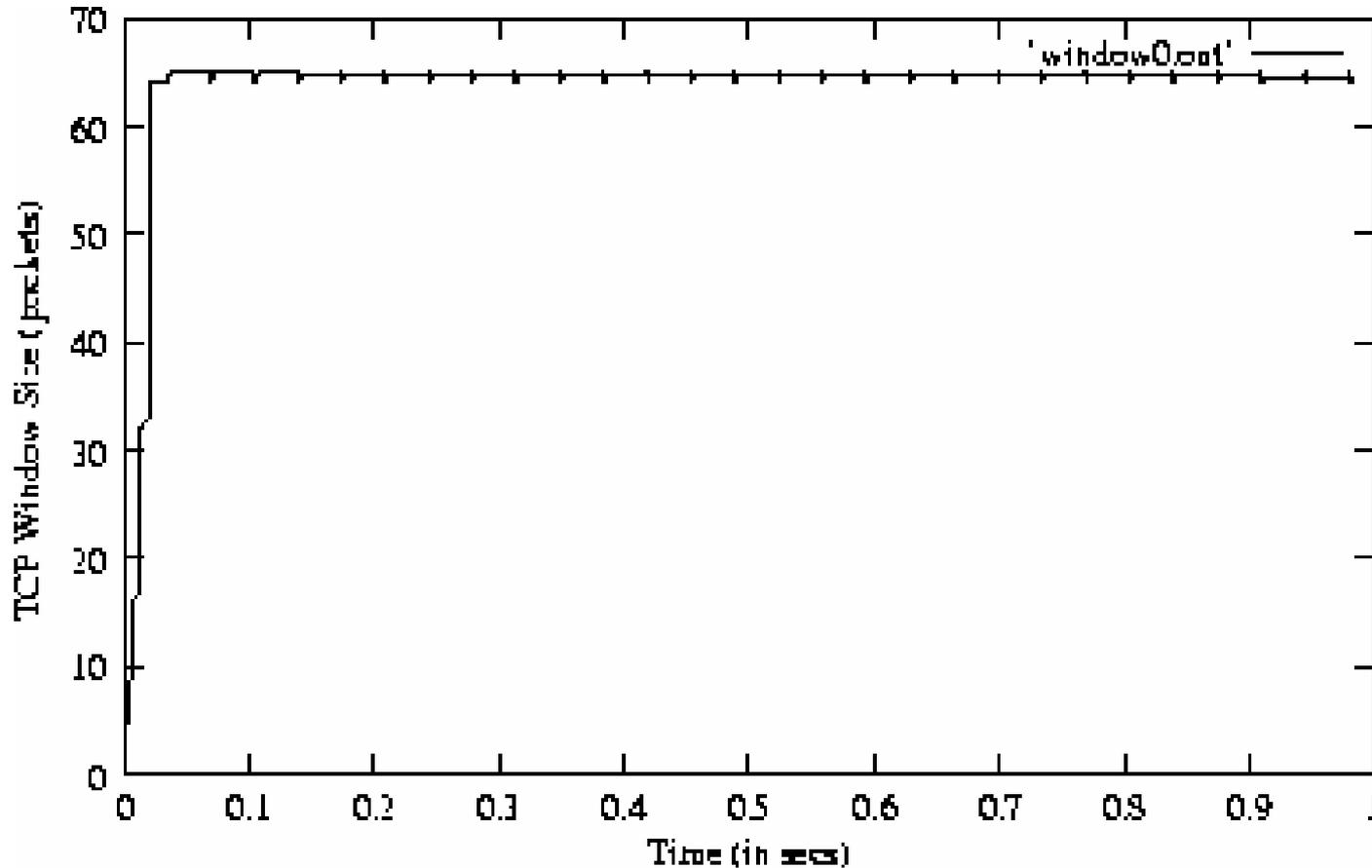


- ❑ Large delay even though total RPR link propagation delay = 1 milliseconds
 - Reflects considerable queueing in the client buffers ($\therefore \uparrow$ convergence time)
- ❑ Oscillations in RTT follow the pattern of aggregate throughput
 - TCP round trip time (Y-axis is Time in seconds) \approx 35 milliseconds



- ❑ Link utilization measured and averaged from time T=0
 - Instantaneous utilization may (will likely) be higher than utilization shown
- ❑ Utilization reached after 1 second = 97.87%

One flow's TCP Window



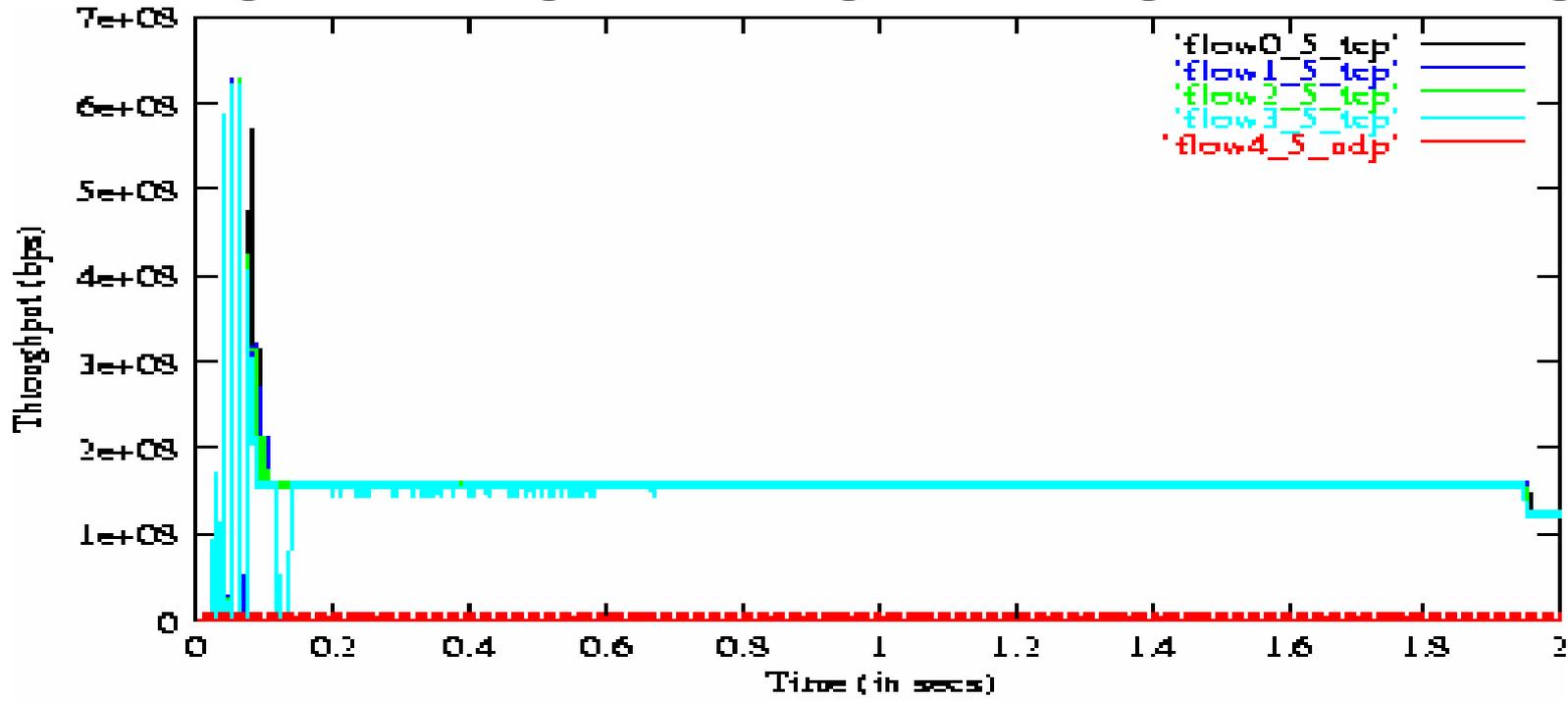
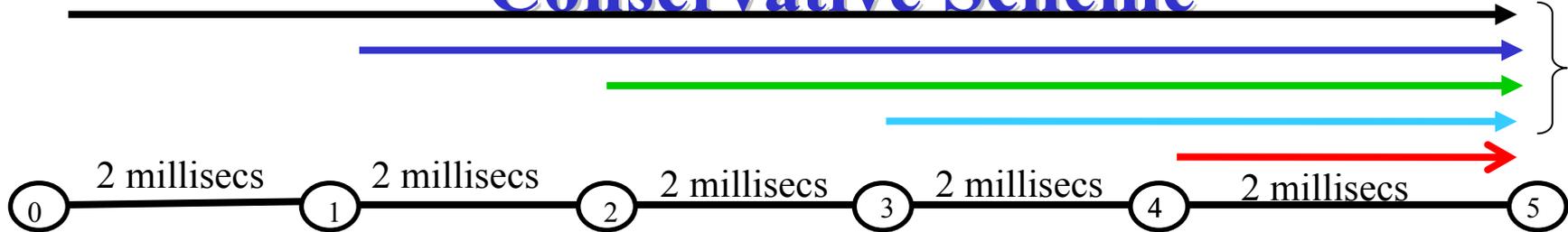
- ❑ TCP window grows reasonably quickly to max. window size of 64



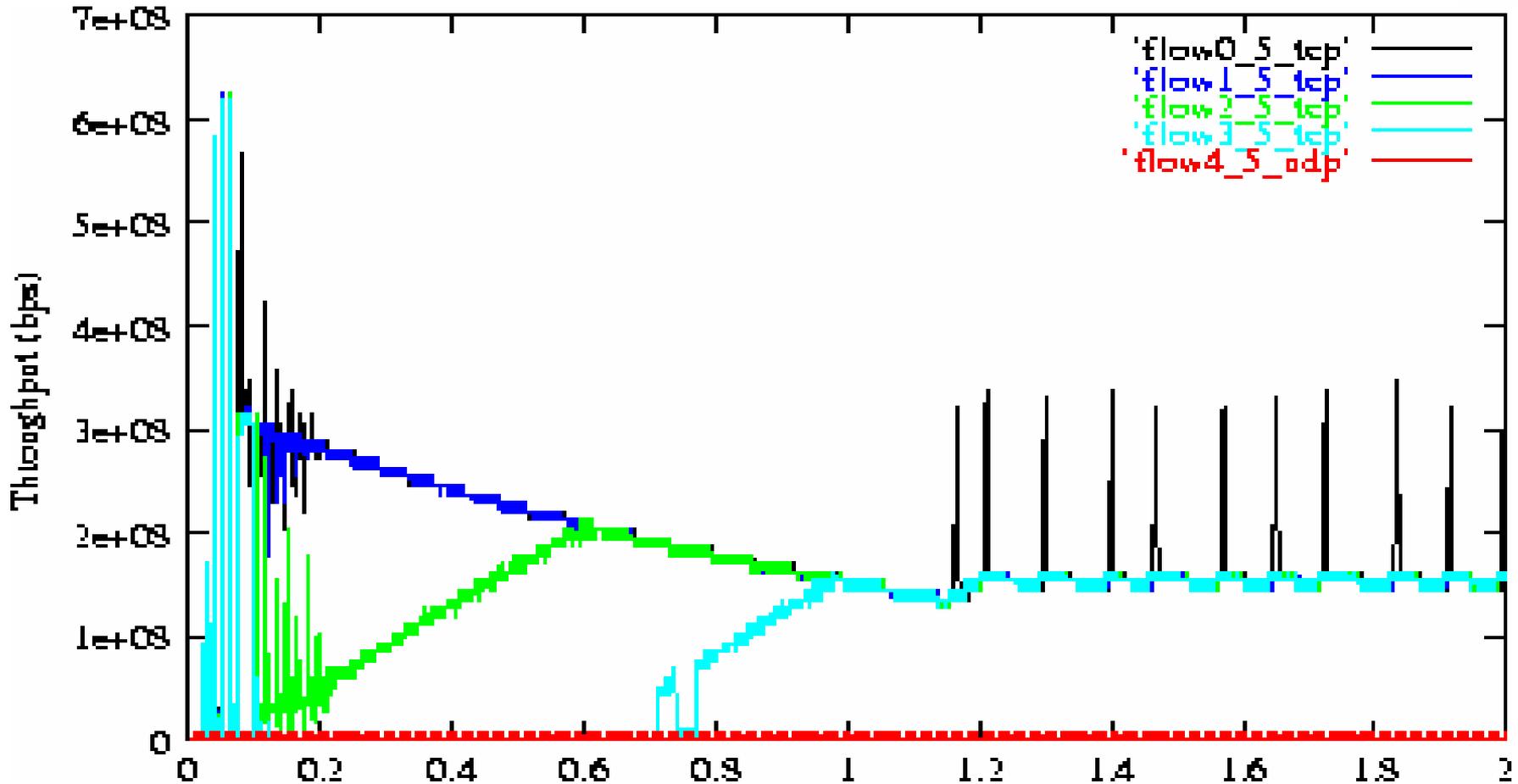
TCP with Large RTT: Enhanced Conservative Scheme



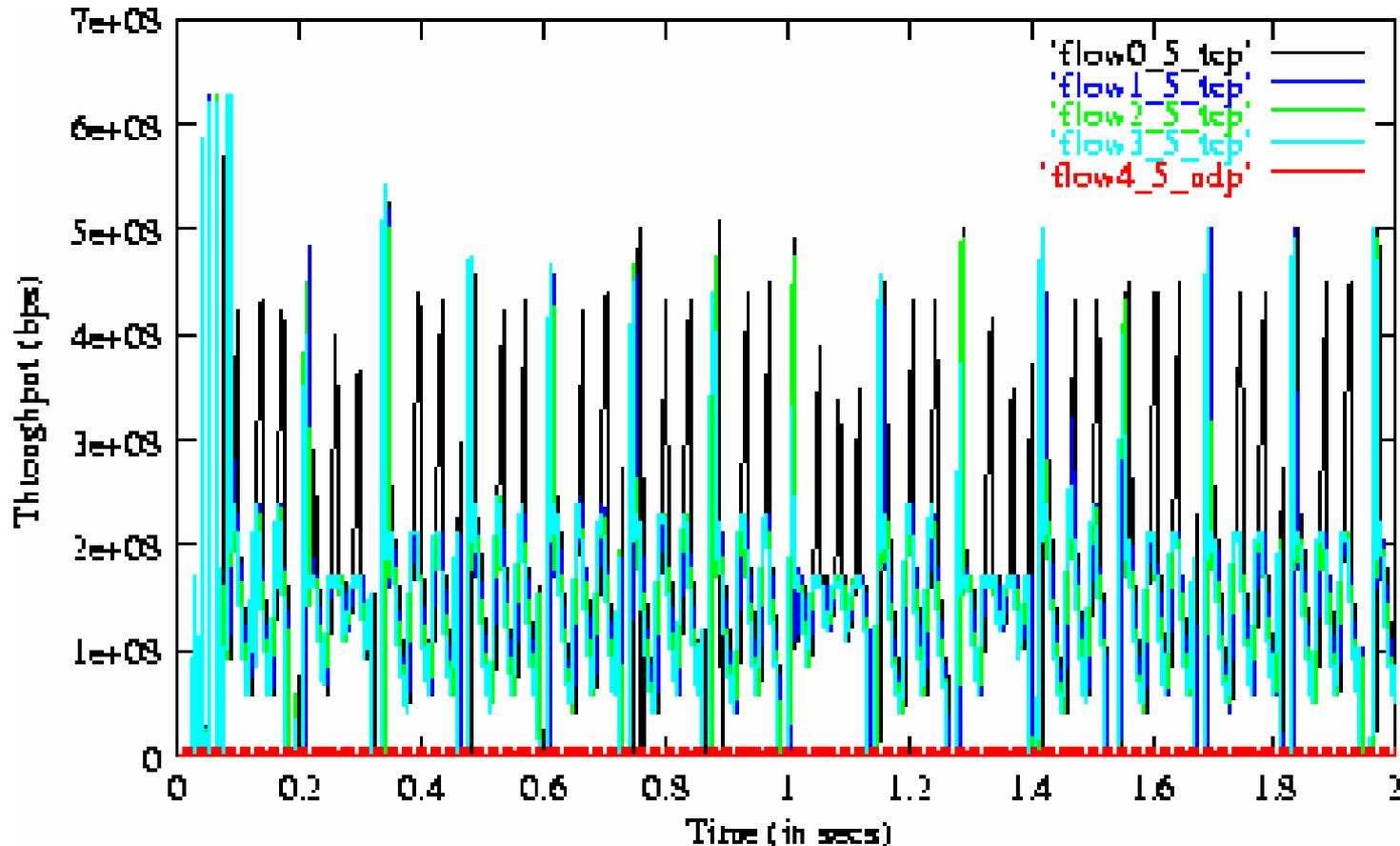
10 Greedy
TCPs each
1Mbps
UDP



- What if we increase the ring size (RTT) considerably?
 - Per hop link propagation delay set to 2 milliseconds (400 Km)
- STQ buffer still maintained at 256 Kbytes
 - Meet difficult challenge to operate with same buffer size over wide dynamic range

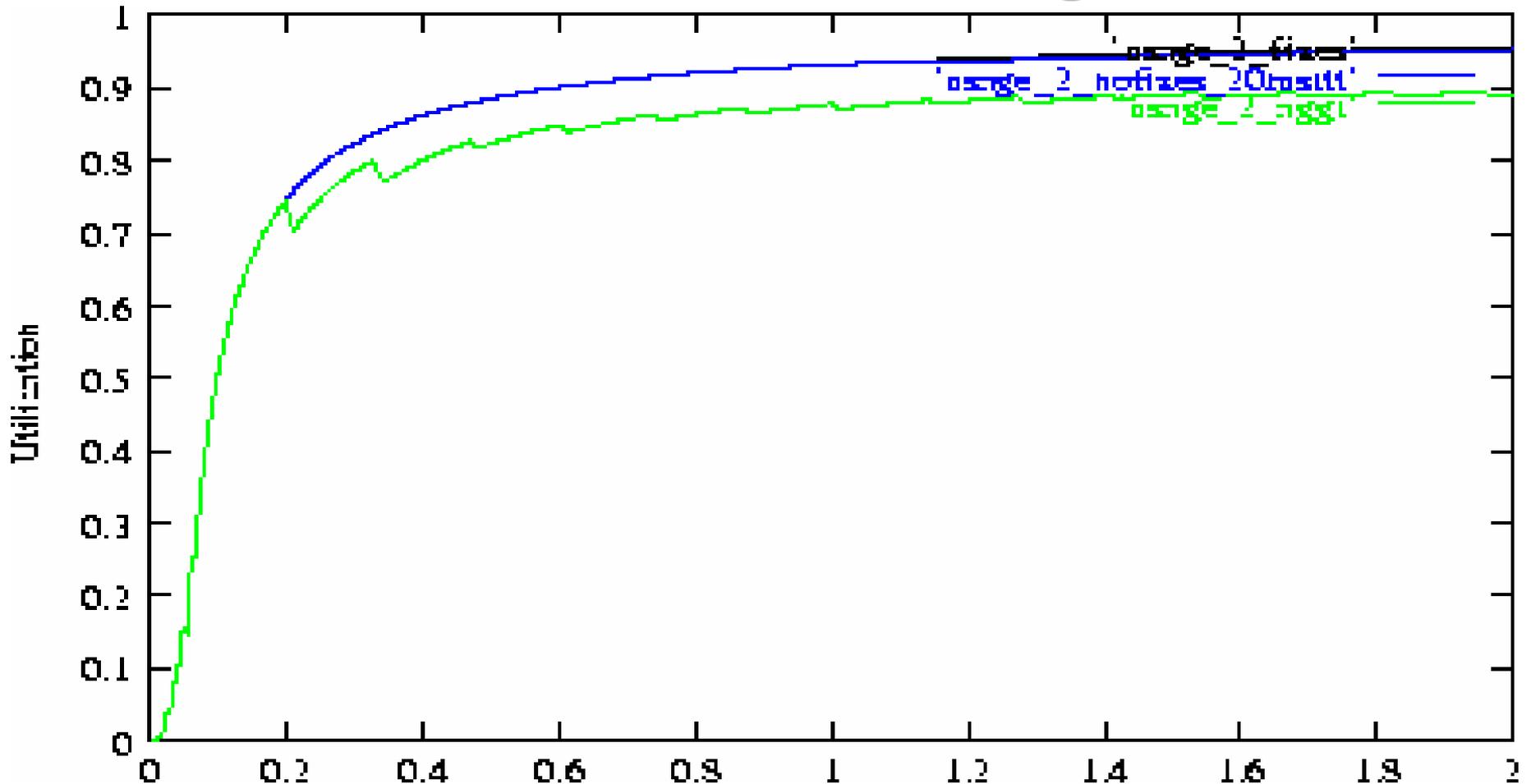


- ❑ Original conservative scheme with fixed RTT estimate = 20 milliseconds
- ❑ Node 0 (farthest upstream) appears to get an unfair advantage
 - Starves downstream nodes (e.g., node 3) when they experience congestion

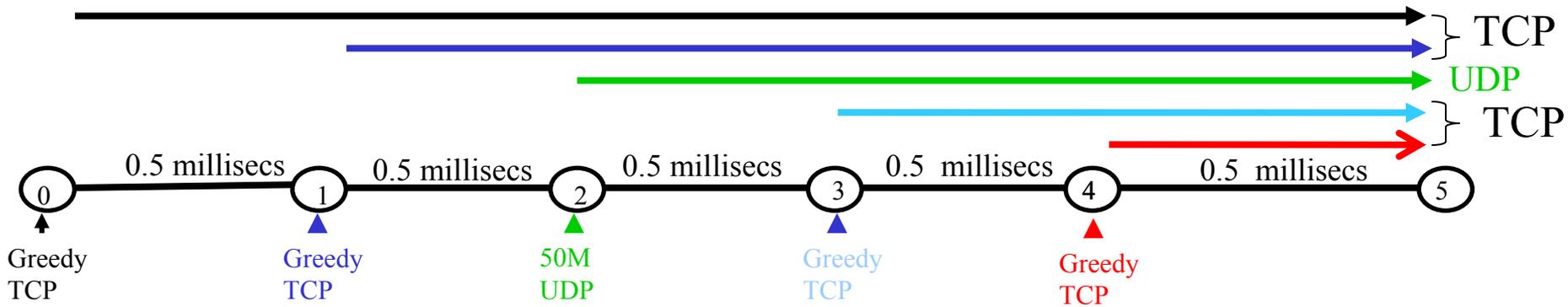


- Aggressive scheme behaves similar to previously explained (Nov. 2002) behavior for large RTTs

Comparison of Bottleneck Link Utilizations with TCP, Large RTT

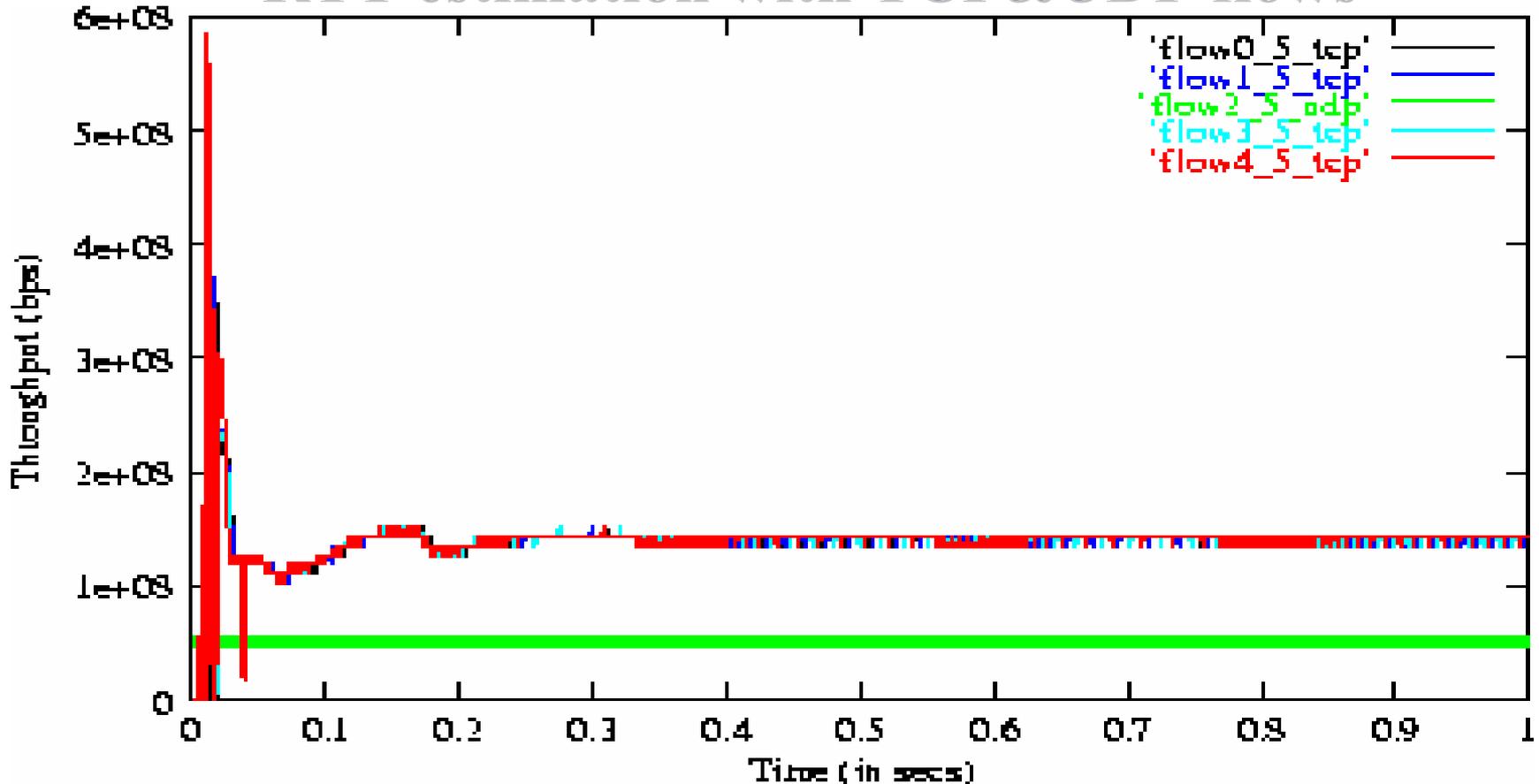


- Link utilization with enhanced conservative scheme is slightly higher than the existing conservative or aggressive schemes

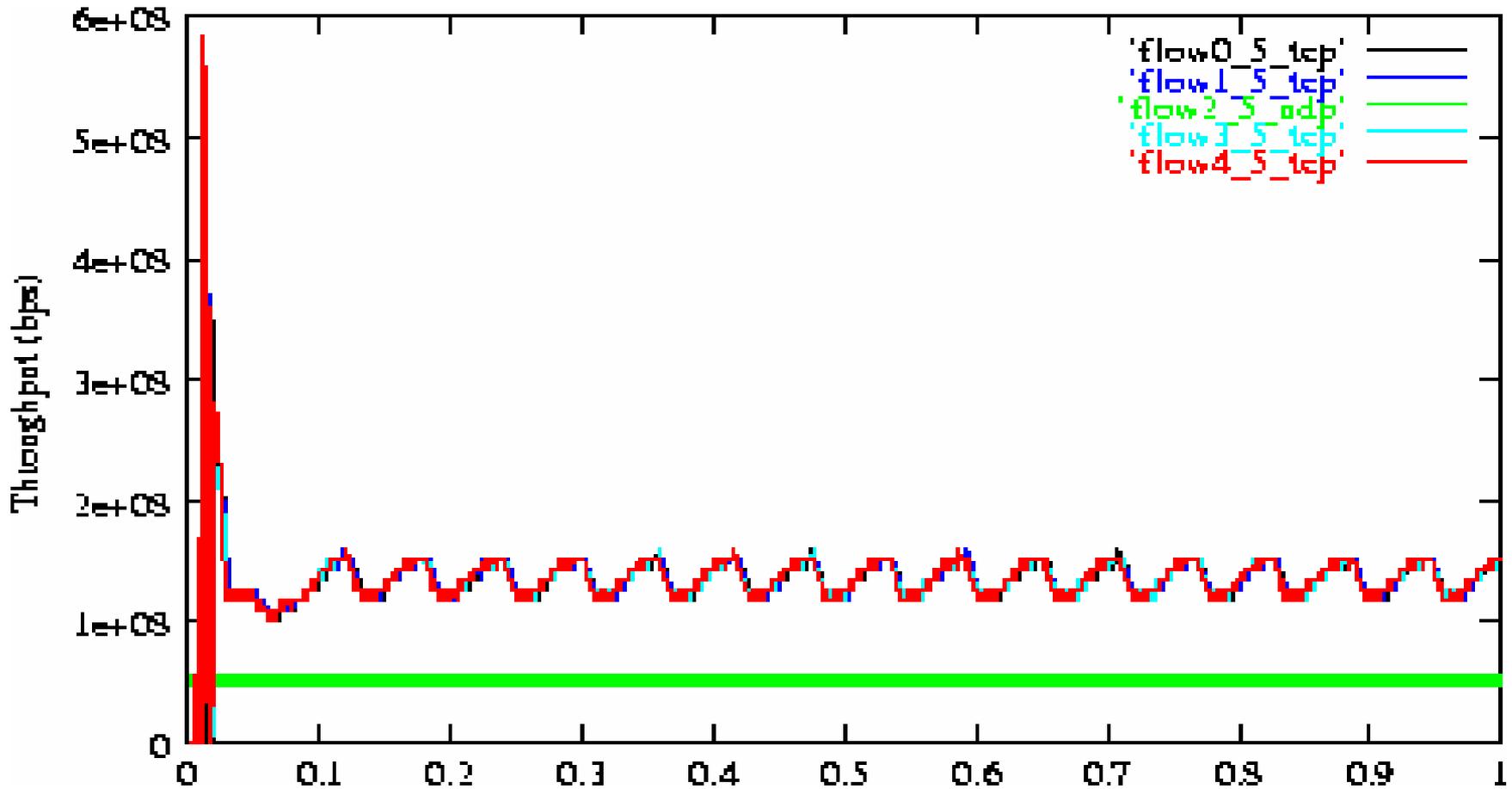


- ❑ What is the benefit of estimating the RTT accurately?
- ❑ We compared the performance of the Enhanced Conservative scheme with 3 experiments.
 - Estimating RTT accurately as suggested here
 - ❖ (5 msec. + STQ delays)
 - Having a fixed RTT = 10% of the round trip propagation delay of domain
 - ❖ $(0.5 * 10) * 0.1 = 0.5$ milliseconds
 - Having a fixed RTT = 20* round trip propagation delay of domain
 - ❖ $(0.5 * 10) * 20 = 100$ milliseconds
- ❑ Simulation configuration uses 4 TCP sources with 10 Greedy TCPs each and 1 fixed rate (50 M) UDP source

Enhanced Conservative Mode with dynamic RTT estimation with TCP&UDP flows

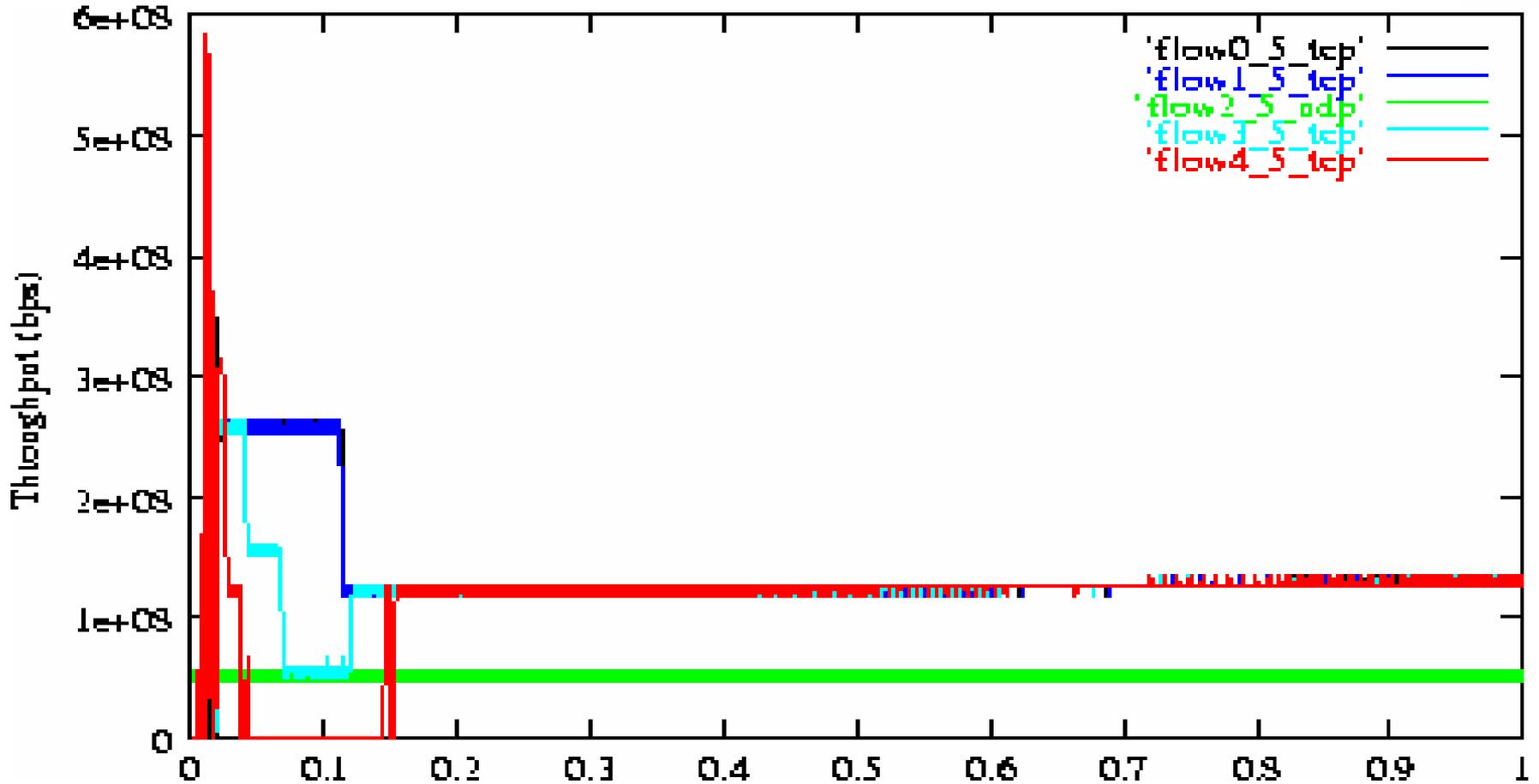


- Estimation of RTT using the protocol specified enables the flows to operate with small oscillations
 - congested node (node 4) does not starve, even during the initial transient
 - approaches fair share (also using node 2's left over bandwidth) over long term



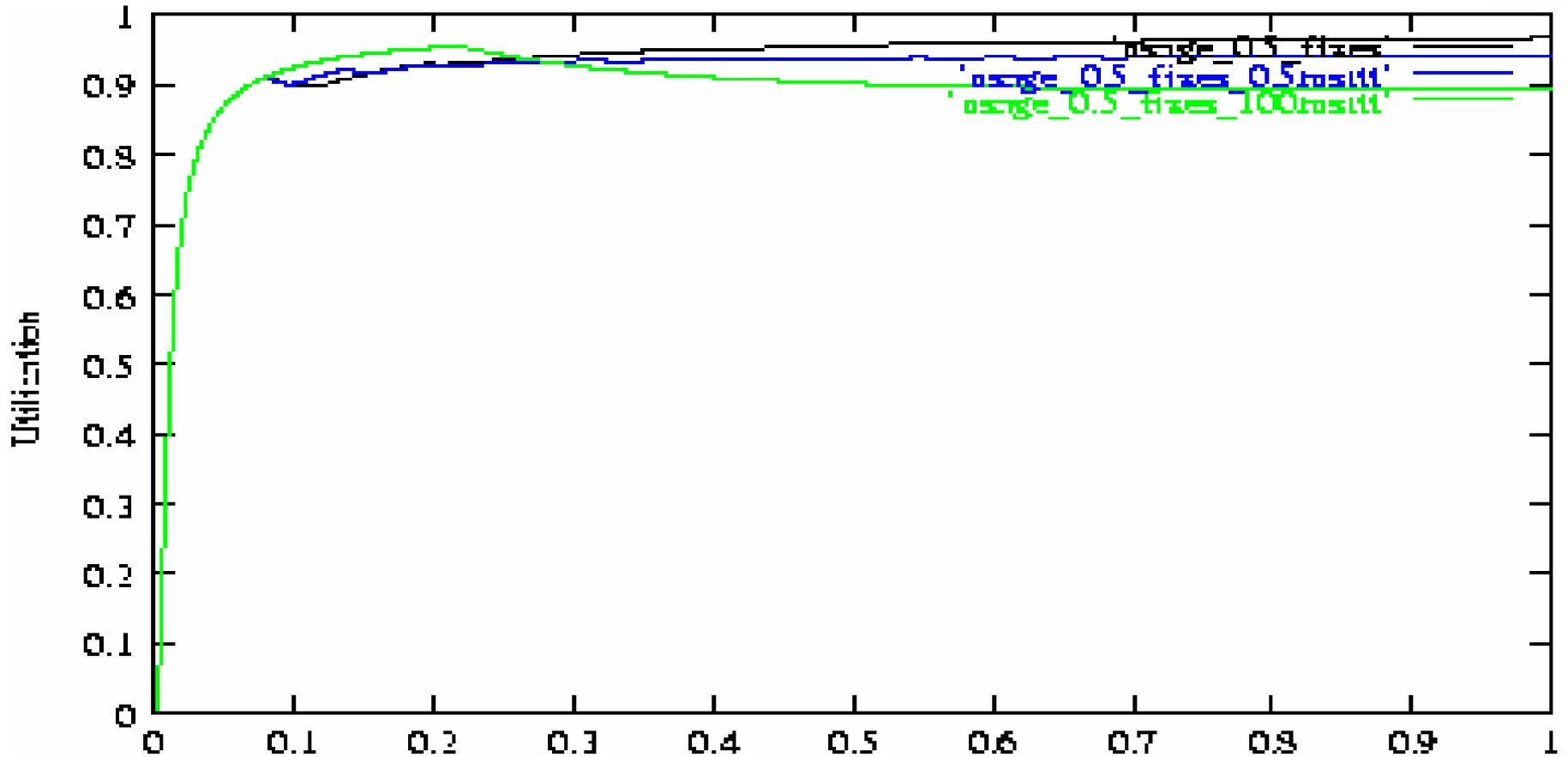
- ❑ Using fixed RTT = 10% round-trip link propagation delays \Rightarrow some oscillation.
- ❑ Average behavior achieves “fair share” (but doesn’t recover unused bandwidth)
 - Reasonable performance for this example

Enhanced Conservative Mode, Fixed RTT=100msec



- ❑ Using fixed RTT = 20 * round-trip link propagation delays ⇒ some oscillation.
- ❑ Node 4 starved for a brief period – as a result of “overdamping”
- ❑ Average behavior achieves “fair share”

Bottleneck Link Utilization



- ❑ Enhanced conservative, dynamic RTT estimate, utilization = 96.73%
- ❑ Enhanced, fixed RTT = 10% of the round trip propagation delay, util= 94.11%
- ❑ Enhanced, fixed RTT = 20* round trip propagation delay, util = 89.45%

Start=0.3ms, stop 0.6 ms

Start=0.2ms, stop 0.5 ms

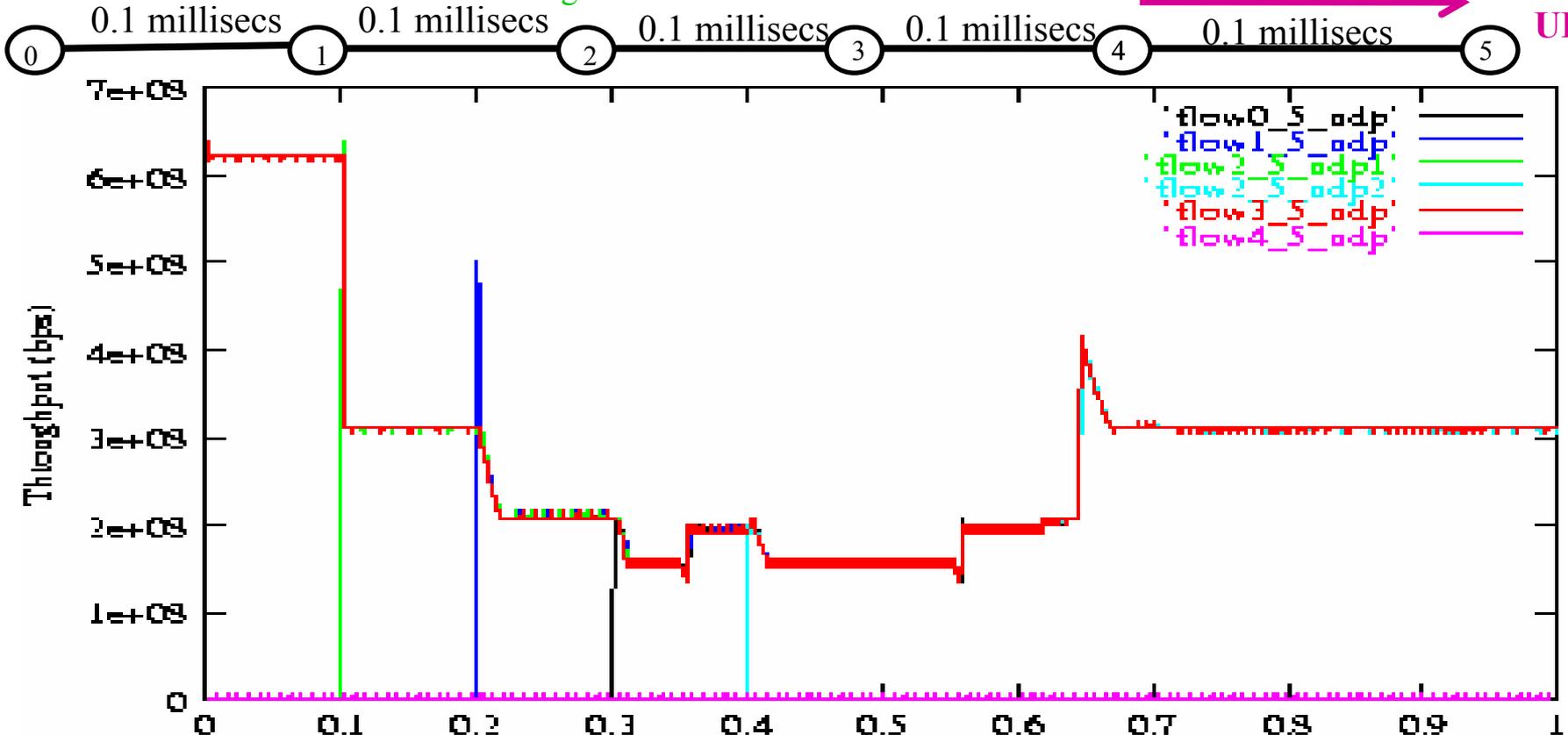
Start=0.1ms, stop 0.3 ms

Start again = 0.4 ms

Start=0

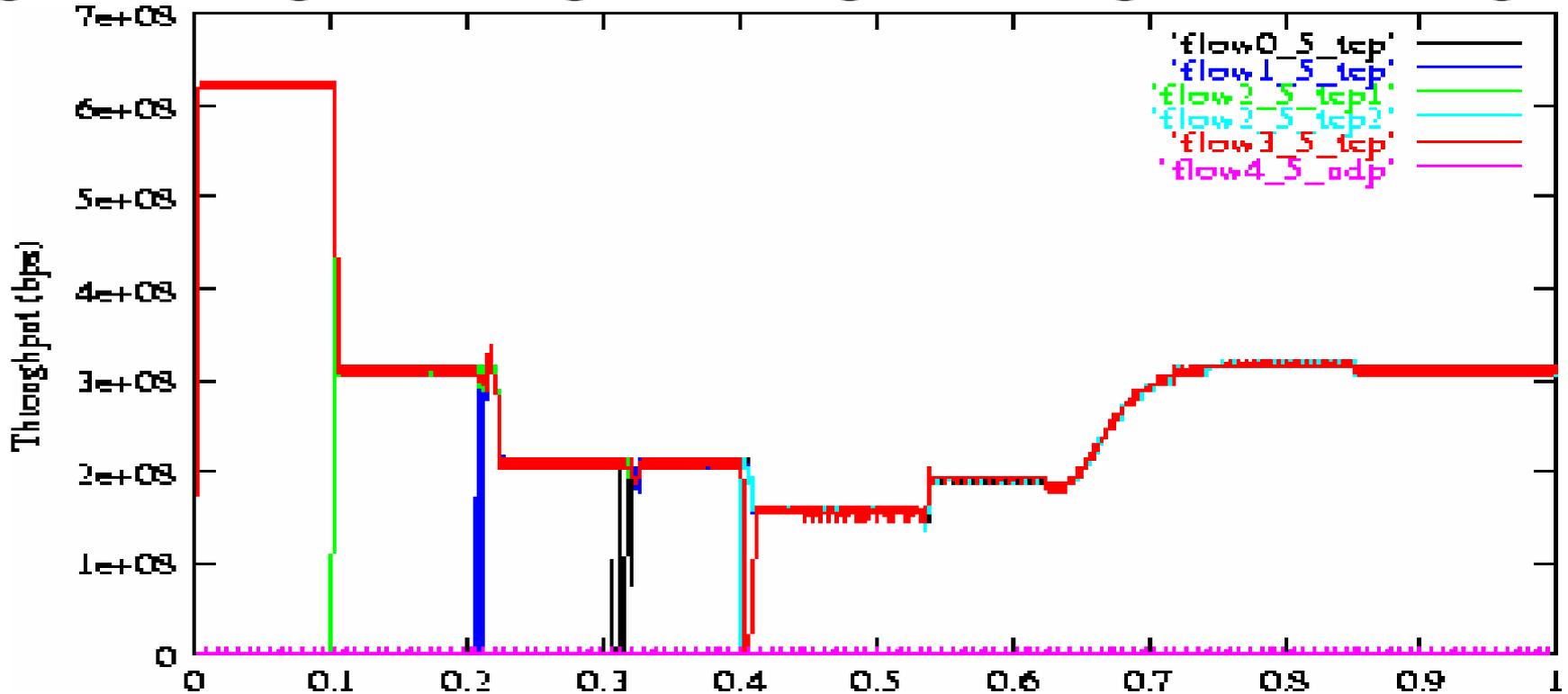
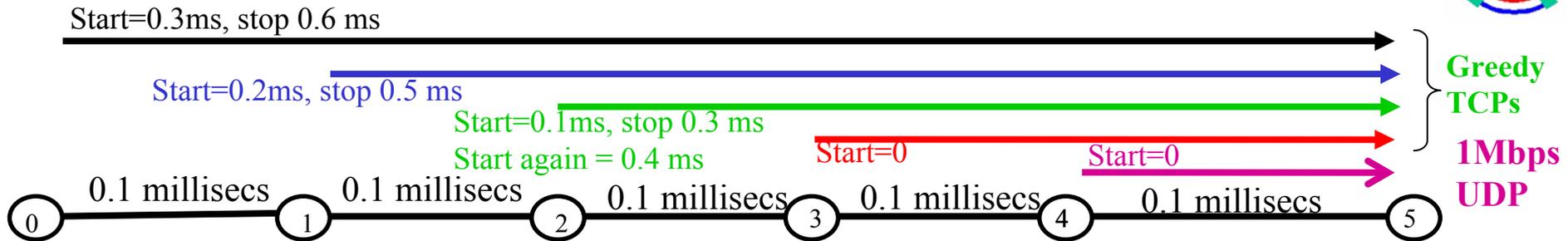
Start=0

Greedy
UDPs
1Mbps
UDP



❑ What happens when we have short lived flows; interaction with long lived flows?

➤ All flows are constant rate (UDP) flows. System is quite responsive to changes



□ TCP flows also respond reasonably

➤ convergence to fair rate when recovering after a flow stops is longer with TCP

- ❑ The conservative fairness allocation scheme for RPR can be made to work reasonably when simple improvements are made
- ❑ We have suggested two broad improvements
 - Improved estimate of LocalFairRate
 - ❖ improved estimate of fair share, account for unused bandwidth, set a lower bound for LocalFairRate when it is reduced during congestion.
 - Estimate Round Trip Time within a Congestion Domain
- ❑ The enhanced conservative mode works quite well
 - improved bottleneck link utilization: very high 90s (%) range
 - operates with a small STQ (256Kbyte) over wide range of ring RTT (2msec)
 - minimizes oscillation; reduces likelihood of starvation of a congested node
 - ❖ A station's transmit rate stabilizes quickly to fair share
 - reasonable convergence times: 3-4 round trips for UDP
- ❑ We encourage/request others to further examine scheme
 - Rate based schemes notorious for complexity and counter-intuitive behavior