

# **Forward Explicit Congestion Notification (FECN) for Datacenter Ethernet Networks**

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These slides are also available on-line at  
<http://www.cse.wustl.edu/~jain/ieee/fecn703.htm>





- ❑ Top 10 Requirements for a Good Scheme
- ❑ FECN Overview
- ❑ Switch Algorithm and Enhancements
- ❑ Simulation Results
  - ❑ FECN with TCP flows
  - ❑ Symmetric Topology
  - ❑ Large Topology
  - ❑ Bursty Traffic



# Datacenter Networks

- ❑ Bounded delay-bandwidth product
  - ❑ High-speed: 10 Gbps
  - ❑ Short round-trip delays
  - ❑ 1 Mb to 5 Mb delay-bandwidth product
- ❑ Storage Traffic  $\Rightarrow$  short access times  $\Rightarrow$  Low delay
- ❑ Packet loss  $\Rightarrow$  Long timeouts  $\Rightarrow$  Not desirable

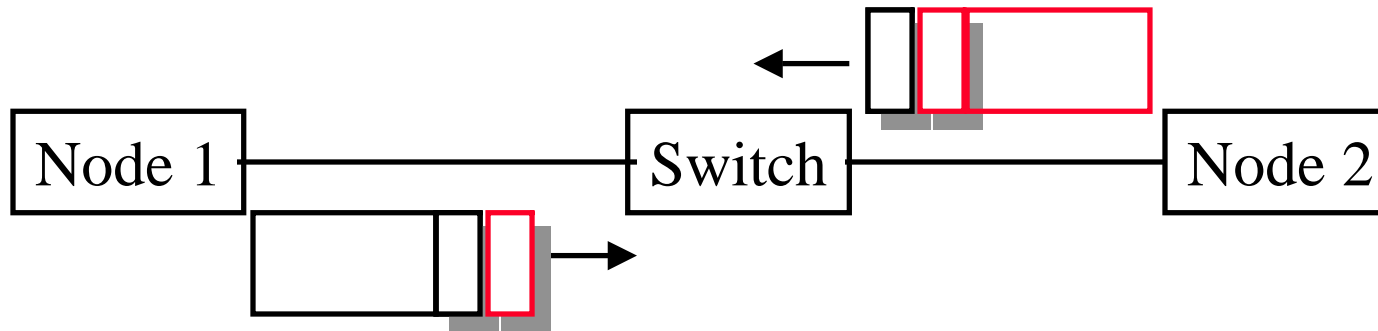


# Top 10 Requirements for a Good Scheme

1. Fast convergence to stability in rates  
Stable rates  $\Rightarrow$  TCP Friendly (IETF feedback)
2. Fast convergence to fairness
3. Good for bursty traffic  $\Rightarrow$  Fast convergence
4. Efficient operation: minimize unused capacity. Minimize chances of switch  $Q=0$  when sources have traffic to send
5. Extremely low (or zero) loss
6. Predictable performance: No local minima
7. Easy to deploy  $\Rightarrow$  Small number of parameters
8. Easy to set parameters
9. Parameters applicable to a wide range of network configurations link speeds, traffic types, number of sources.
10. Applicable to a variety of switch architectures and queueing/scheduling disciplines



# FECN Overview



- ❑ Periodically, the sources piggyback a “Rate Discovery Tag” (RD tag) on the outgoing packet.
- ❑ The tag contain only rate, Rate limiting Q ID, and direction. (Direction = Forward (discovery) tag or Returning tag)
- ❑ The sender initializes the RD tag with  $\text{rate} = -1$  ( $\Rightarrow \infty$ )
- ❑ The switches adjust the rate down if necessary
- ❑ The receiver copies the forward RD tag in a control packets in the reverse direction
- ❑ Source adjusts to the rate received

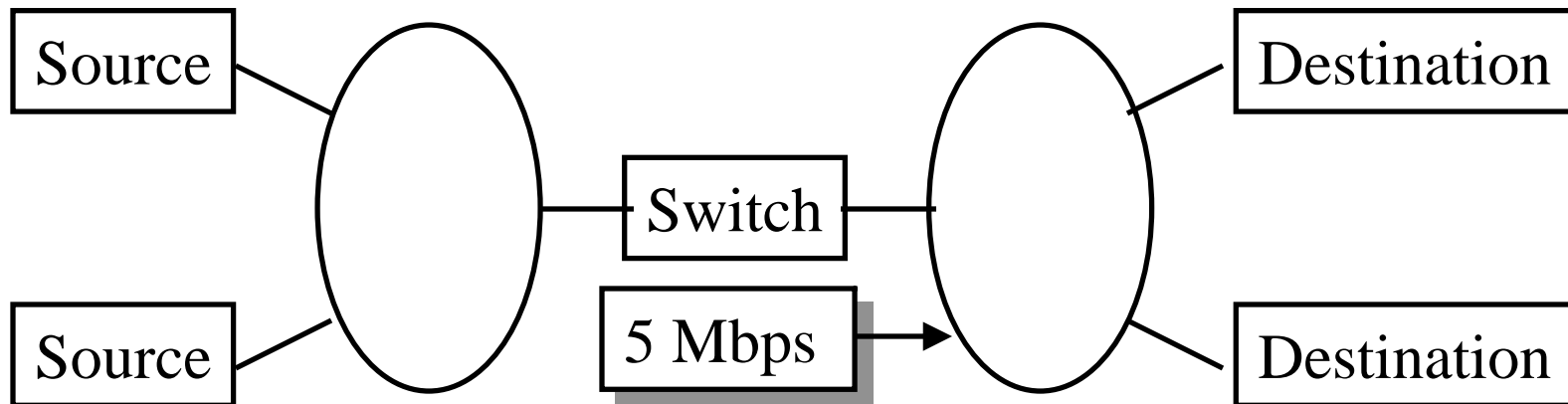


# FECN: Observations

- ❑ This is similar to what is done in TCP/IP, Frame Relay, ATM with 1 bit in every packet ( $n=1$ ).
- ❑ ATM ABR had a similar explicit rate indication that was selected after 1 year of intense debate and scrutiny.
- ❑ Only the feedback format has to be standardized
- ❑ No need to standardize switch algorithm.
- ❑ Vendor differentiation: Different switch algorithms will “inter-operate” although some algorithms will be more efficient, more fair, and achieve efficiency/fairness faster than others.
- ❑ We present a sample switch algorithm and show that it achieves excellent performance.



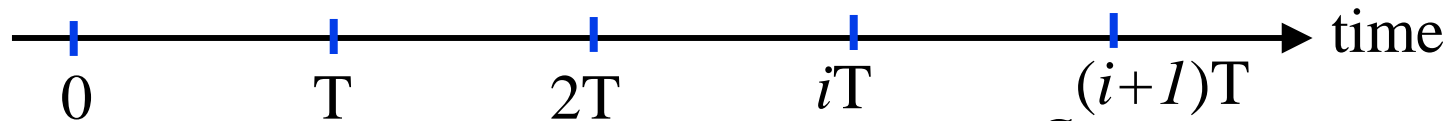
# Switch Algorithm



- ❑ The switch use the same “**Advertised Rate**” in all RD tags
- ❑ All sources passing through the switch get the same feedback.
- ❑ The sources send at the rate received.



# The Basic Switch Algorithm



0. Start with an Advertised Rate of  $r$ .  $r_0 = \frac{C}{N_0}$   
Here  $C$  is the link capacity.
1. Measure input rate every  $T$  interval
2. Compute overload factor  $z$  in the last  $T$  interval
3. Change the advertised rate to  $r/z$
4. In every RD tag: set rate to  $\min\{\text{rate in tag, advertised rate}\}$
5. Go back to step 1

Although this simple algorithm will work but:

- It will oscillate even if the rate is close to optimal.
- Queues will not be constant  $\Rightarrow$  *Need a Q Control Fn*





# Enhancement 1: Queue-Control

1. **Measurement:** Let  $A_i$  be the measured arrival rate in bits/s then the load factor is  $z = A_i/C$ . We update this load factor based on the queue length so that the *effective load factor* is:

$$\rho_i = \frac{z}{f(q_i)} = \frac{A_i}{f(q_i) \times C}$$

2. **Bandwidth Allocation:**

$$r_{i+1} = \frac{r_i}{\rho_i}$$

**Note:** We also tried additive queue control. It has similar performance.

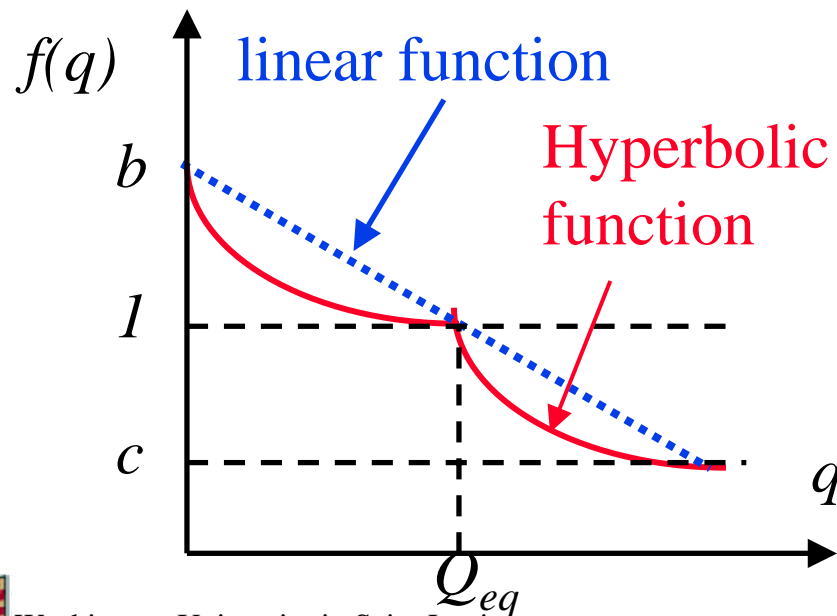


# Queue Control Function: $f(q)$

**Idea:** Give less rate if queue length is large and more if queue length is small compared to desired queue length of  $Q_{eq}$  and  $f(Q_{eq})=1$

$$f(q) = \begin{cases} \geq 1 & q \leq Q_{eq} \\ = 1 & q = Q_{eq} \\ \leq 1 & q \geq Q_{eq} \end{cases}$$

Reserves some capacity for draining the queue.



We analyzed many different functions and recommend the hyperbolic function because it gives smaller oscillations. [See reference]



# Queue Control Function (Cont)

- **Linear Function:**  $k$  is some constant

$$f(q) = 1 - k \frac{q - Q_{eq}}{Q_{eq}}$$

- **Hyperbolic function:**  $a, b, c$  are constants. Pre-computed in a table.

$$f(q) = \begin{cases} \frac{bQ_{eq}}{(b-1)q + Q_{eq}}, & \text{if } q \leq Q_{eq}; \\ \max \left( c, \frac{aQ_{eq}}{(a-1)q + Q_{eq}} \right), & \text{otherwise.} \end{cases}$$

q	f(q)

In all simulations,  $a = 1.1, b = 1.002, c = 0.1$



# Enhancement 2: Exponential Averaging

Exponentially weighted average in the Switch:

$$r_i = \alpha \frac{r_{i-1}}{\rho_i} + (1 - \alpha)r_{i-2}$$

$$\alpha \in (0, 1)$$

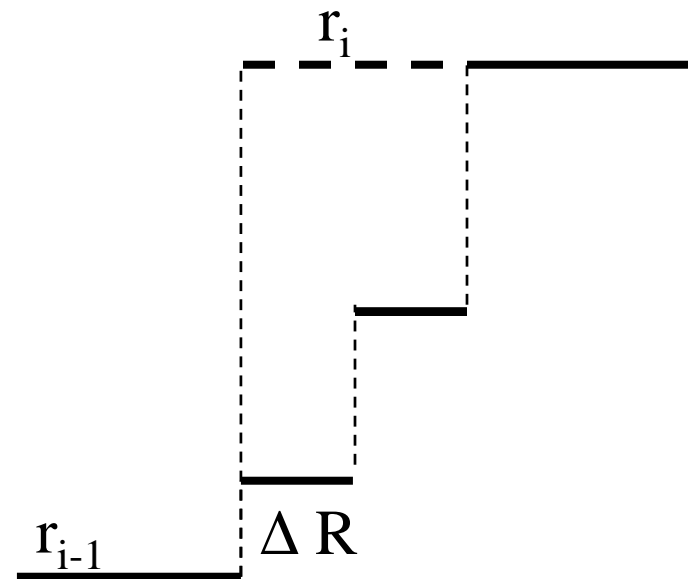
Remembers recent history. In all simulations  $\alpha = 0.5$



# Enhancement 3: Limited Rate Increase

## Limit rate increase in the switch

```
IF ( $q < Q_{eq}$ ) THEN
     $\Delta R = 1.414\Delta R$ 
ELSE IF ( $q > Q_{sc}$ ) THEN
     $\Delta R = 0.707\Delta R$ 
END IF
IF ( $r_i - r_{i-1} > \Delta R$ ) THEN
     $r_i = r_{i-1} + \Delta R$ 
END IF
```



- **Strategy:** Take small jumps. Jump size increases with every step. Results in fast rise time but avoids sudden queue increase if false signal.



## Enhancement 4: Variable Capacity Adjustment

- If capacity of the link reduces due to failure of a component link in an aggregated link or other reasons, the allocated rate is reduced accordingly.

IF ( $c_i < c_{i-1}$ ) THEN

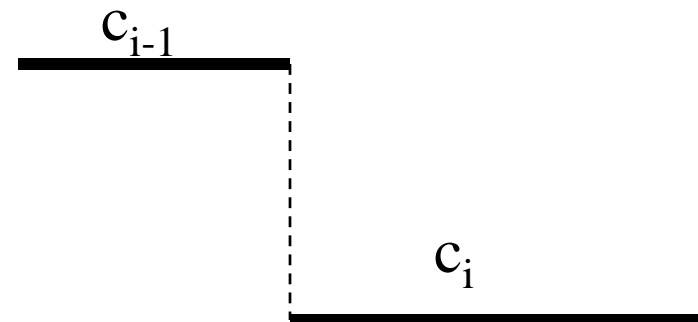
$$r_i = (c_i/c_{i-1})r_i$$

$$r_{i-1} = (c_i/c_{i-1})r_{i-1}$$

END IF

$$r_{i-2} = r_{i-1}$$

$$r_{i-1} = r_i$$



# Enhancement 5: Time Based Sampling

**Time-based sampling at the source:** Packet tagged if time since the last time tag was sent is more than  $\tau$

In all simulations  $\tau = T$



# General Simulation Parameters

- ❑ Queue control function: Hyperbolic
- ❑ Packet size = 1500 B
- ❑ Measurement interval  $T = 1$  ms





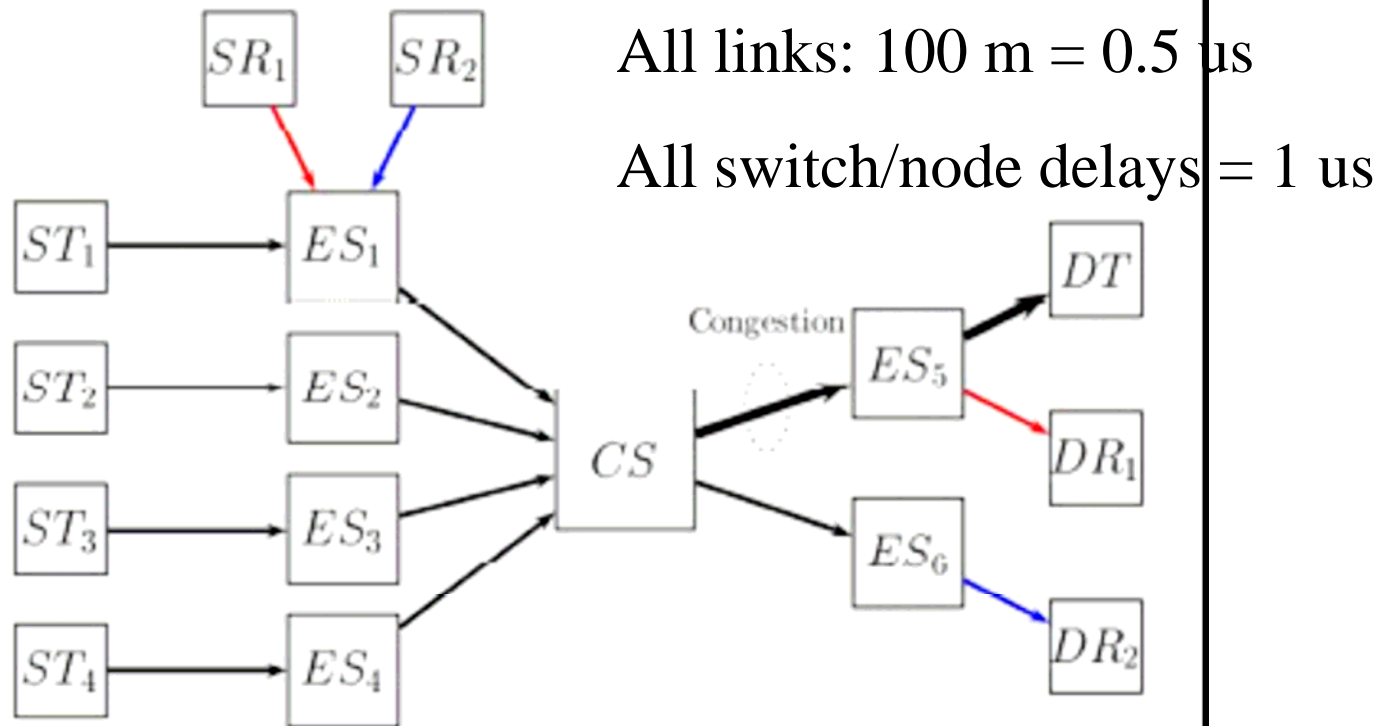
# Baseline Simulation Results

1. FECN with TCP flows
2. Symmetric Topology
3. **Large** Topology with 100 flows
4. **Bursty Traffic**: Pareto-distributed burst time
5. Output-Generated **Hot-Spot** Scenario
6. Output-Generated Hot-Spot Scenario with **long delay**

**All simulations use the same parameter values!**



# FECN with TCP flows



- ❑ 6-source topology
- ❑ SR1-to-DR1 and SR2-to-DR2 are *reference flows*
- ❑ SR<sub>i</sub>-to-DT are four flows that share the bottleneck link



## FECN with TCP flows (Cont)

- $T = 1$  ms
- Total simulation time = 1 sec
- Workload
  - ST1-ST4: 10 parallel TCP connections transferring 1 MB each continuously  
1 Transaction = 1 MB transfer
  - Reference flows: 1 TCP connection transferring 10kB each with average idle time of 16 us for SR1 and 1 us for SR2



# Simulation Results

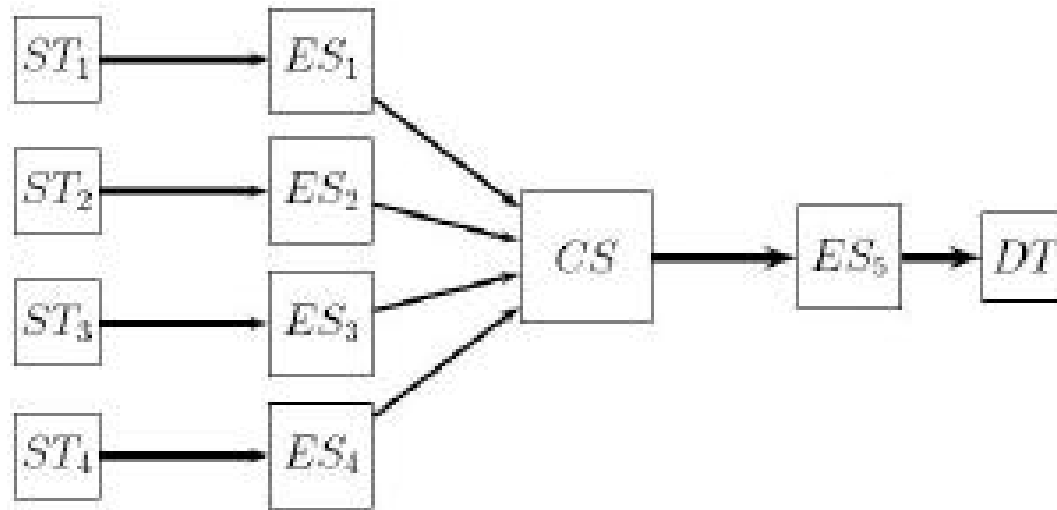
	Reference Flow 1			Reference Flow 2		
Congestion Mechanism	Throughput (Transactions/s)	Throughput (Gbps)	Transaction Completion Time (us)	Throughput (Transactions/s)	Throughput (Gbps)	Transaction Completion Time (us)
None	556	0.06	1780.78	16634	1.44	59.11
FECN	6970	0.604	127.63	16630	1.44	59.16

Congestion Mechanism	Average Throughput (Gbps)	Jain Fairness Index	Link Utilization (%)
None	2.49	2%	99.9
FECN	2.35	99%	99.9

**Conclusions:** FECN can protect fragile TCP flows and improve its goodput and fairness significantly. FECN reduced packet loss from 50,008 packets to 0 packets.



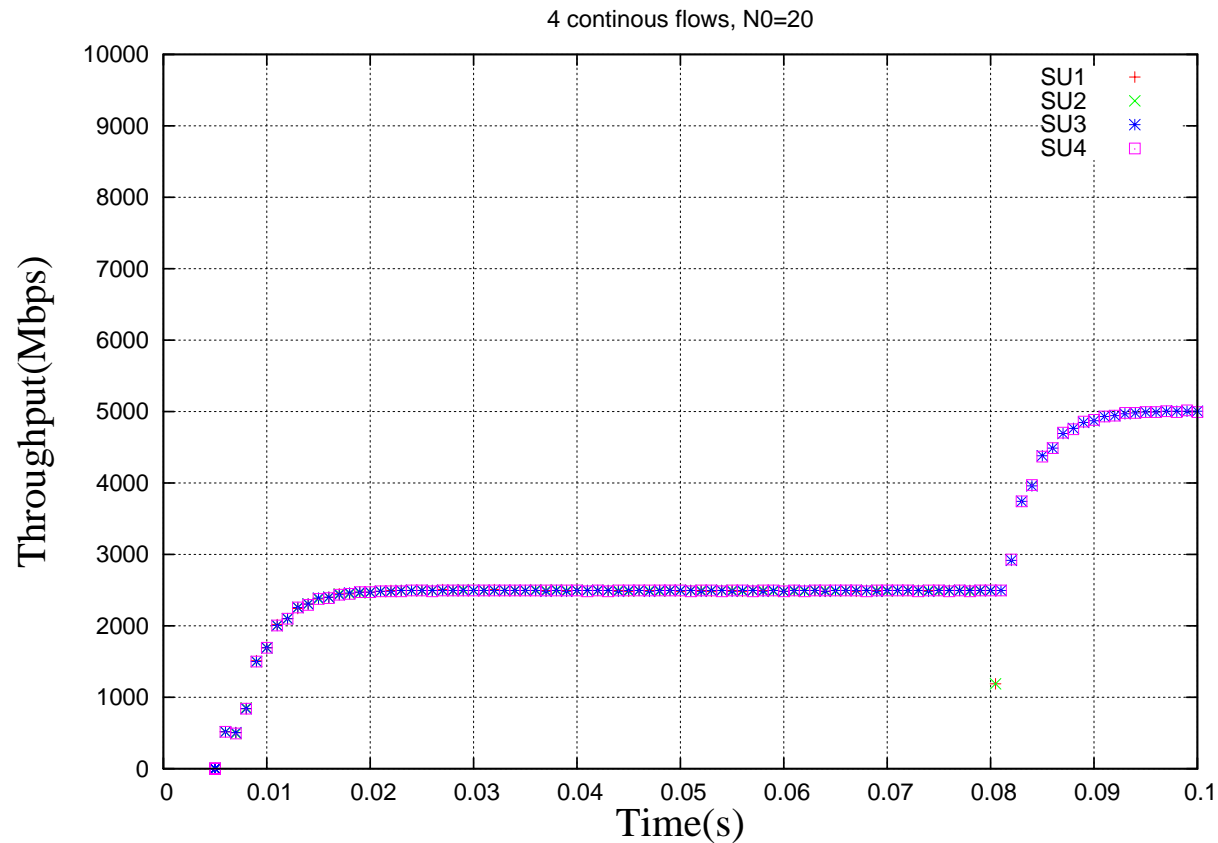
# Symmetric Topology: Configuration



- ❑ UDP *Bernoulli Traffic* with average 5 Gbps rate
- ❑ Measurement Interval  $T$  is 1 ms,  $N_0 = 20$
- ❑ Simulation Time is 100 ms, all sources starts at 5 ms
- ❑ At 80 ms, 2 sources stop
- ❑ Per-hop-delay=0.5 us, switch/node delay is 1 us



# Symmetric Topology: Source Throughput



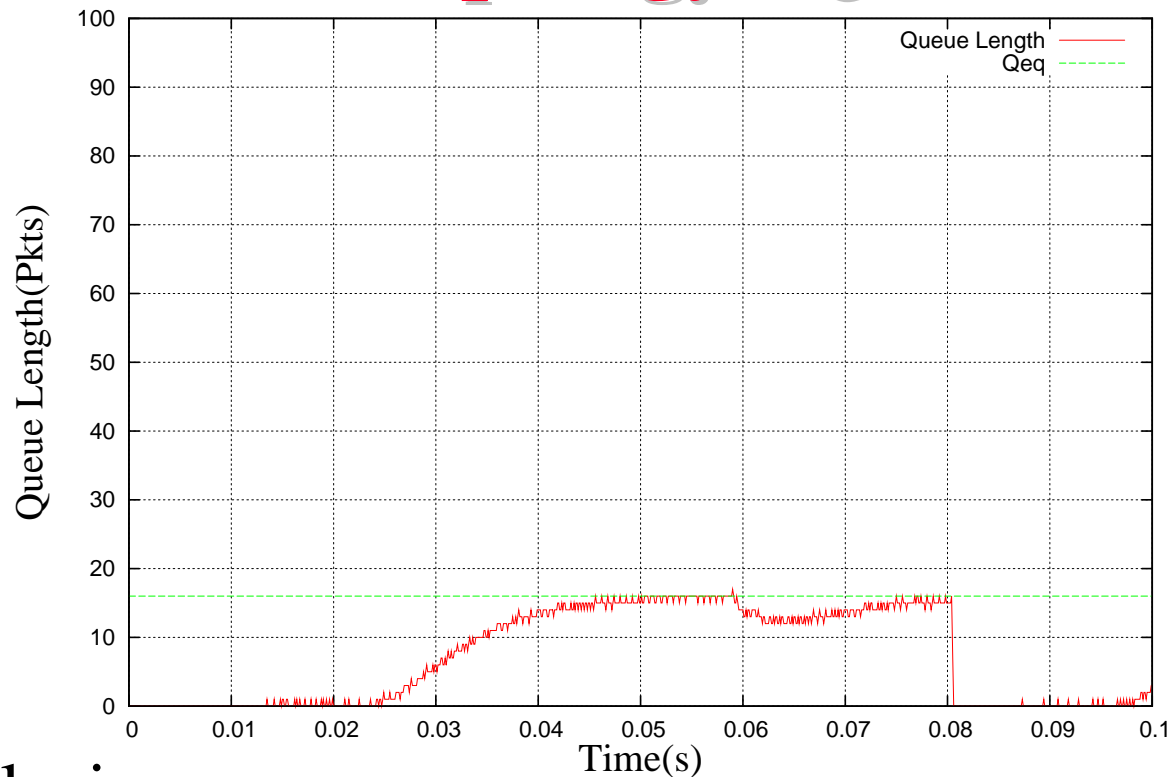
## Conclusions:

- Four sources overlap  $\Rightarrow$  Perfect Fairness!
- Fast Convergence: around 10 ms



# Symmetric Topology: Queue Length

4 continuous flows,  $N_0=20$

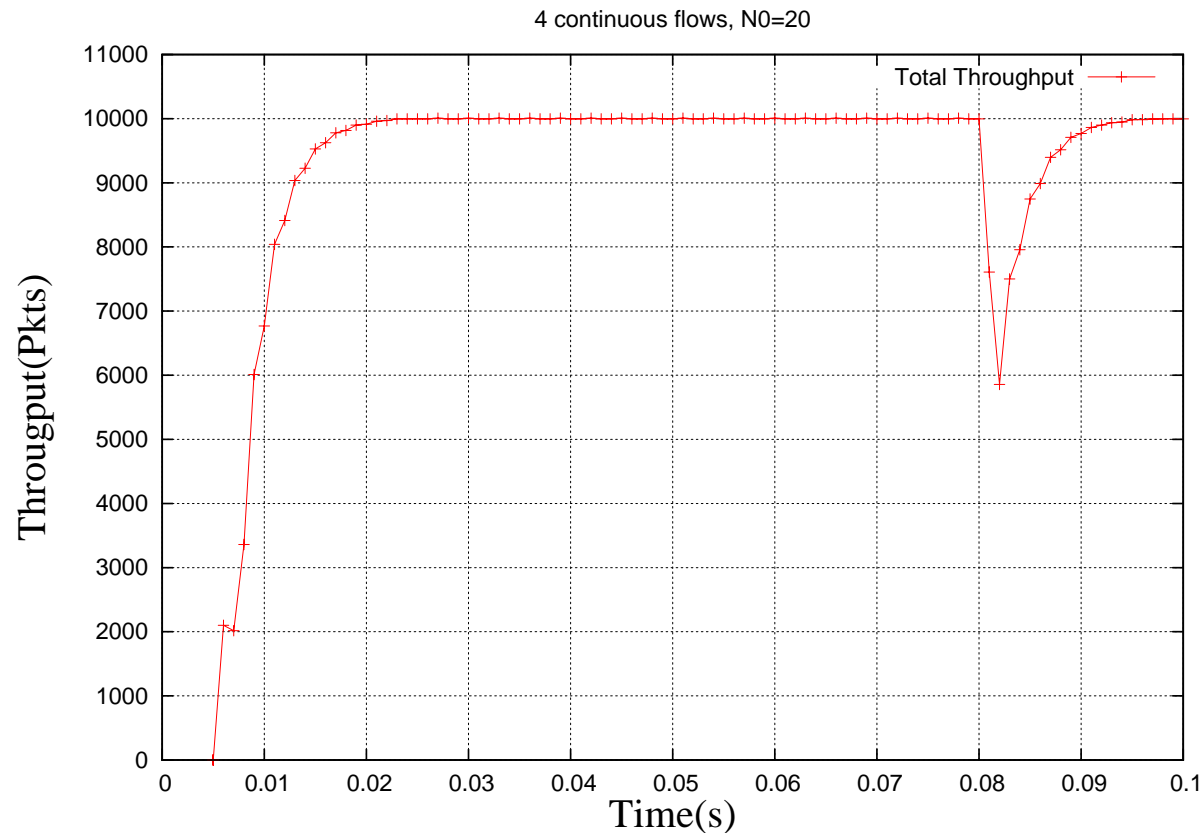


## □ Conclusions:

- Queue builds up to Qeq and can stay there.
- Queue never overflows



# Symmetric Topology: Link Utilization

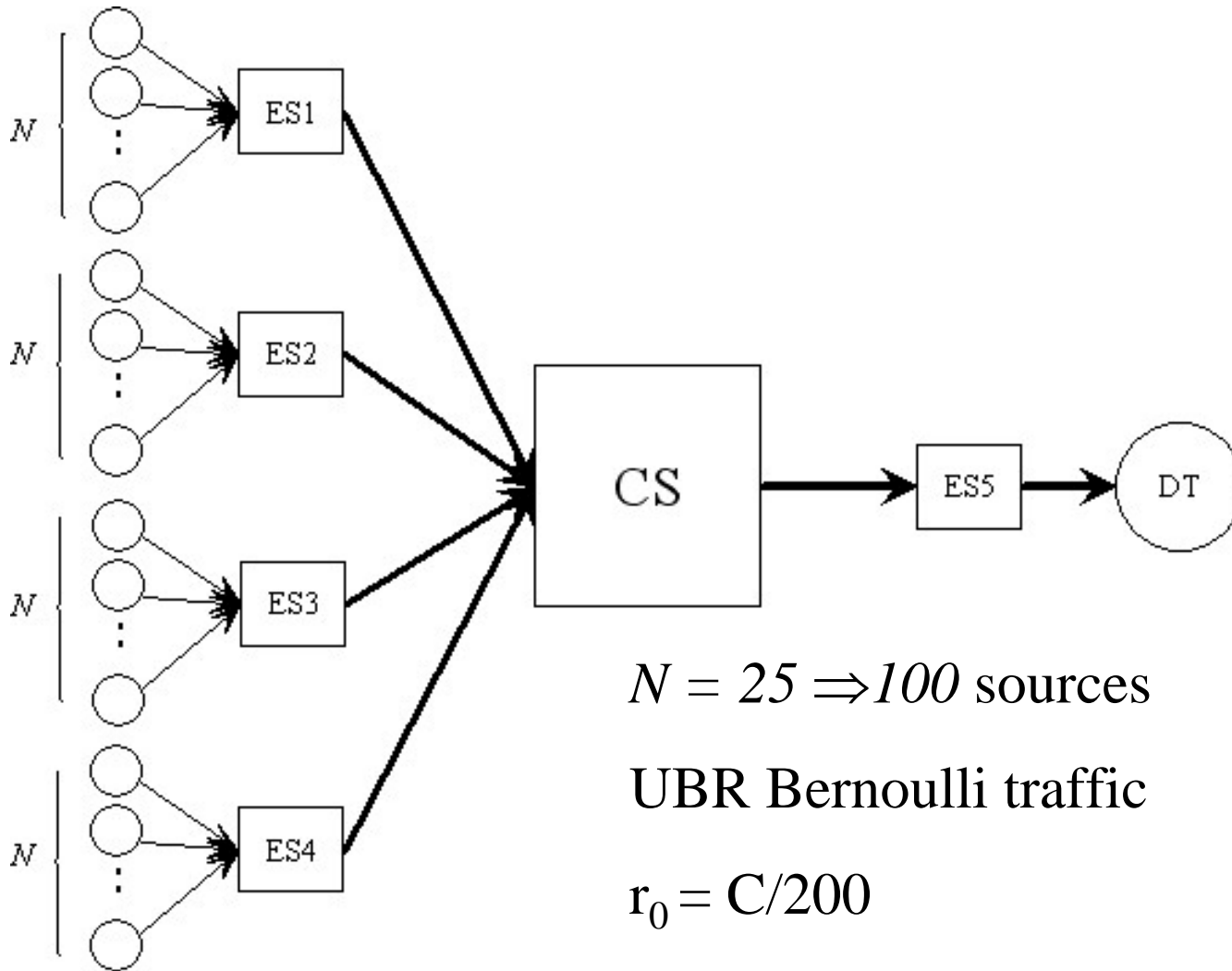


- ❑ **Conclusions:** Link is highly utilized when the rate achieves the fair share in around 10 ms.





# Large Topology: Configuration



$N = 25 \Rightarrow 100$  sources

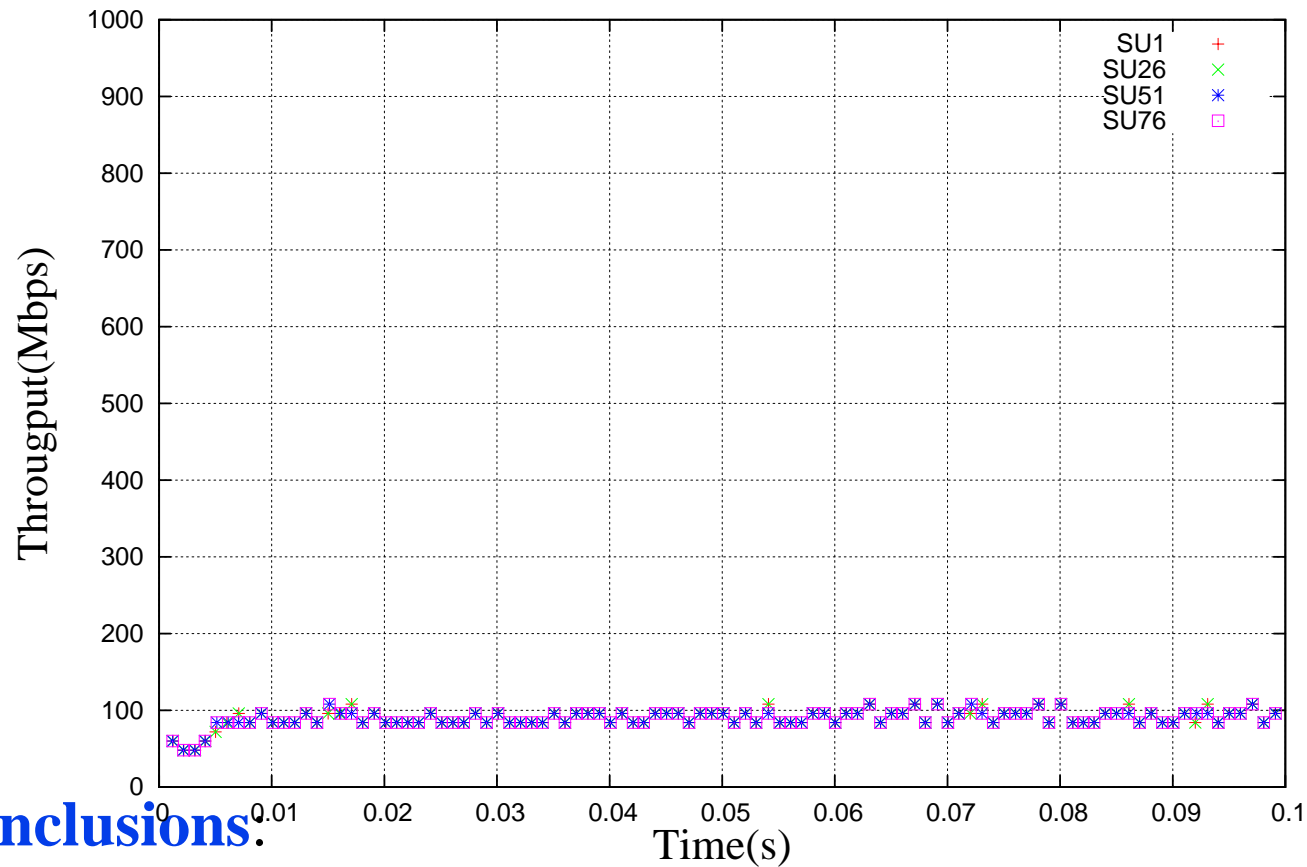
UBR Bernoulli traffic

$r_0 = C/200$



# Large Topology: Source Rates

100 continuous flows,  $N_0=200$

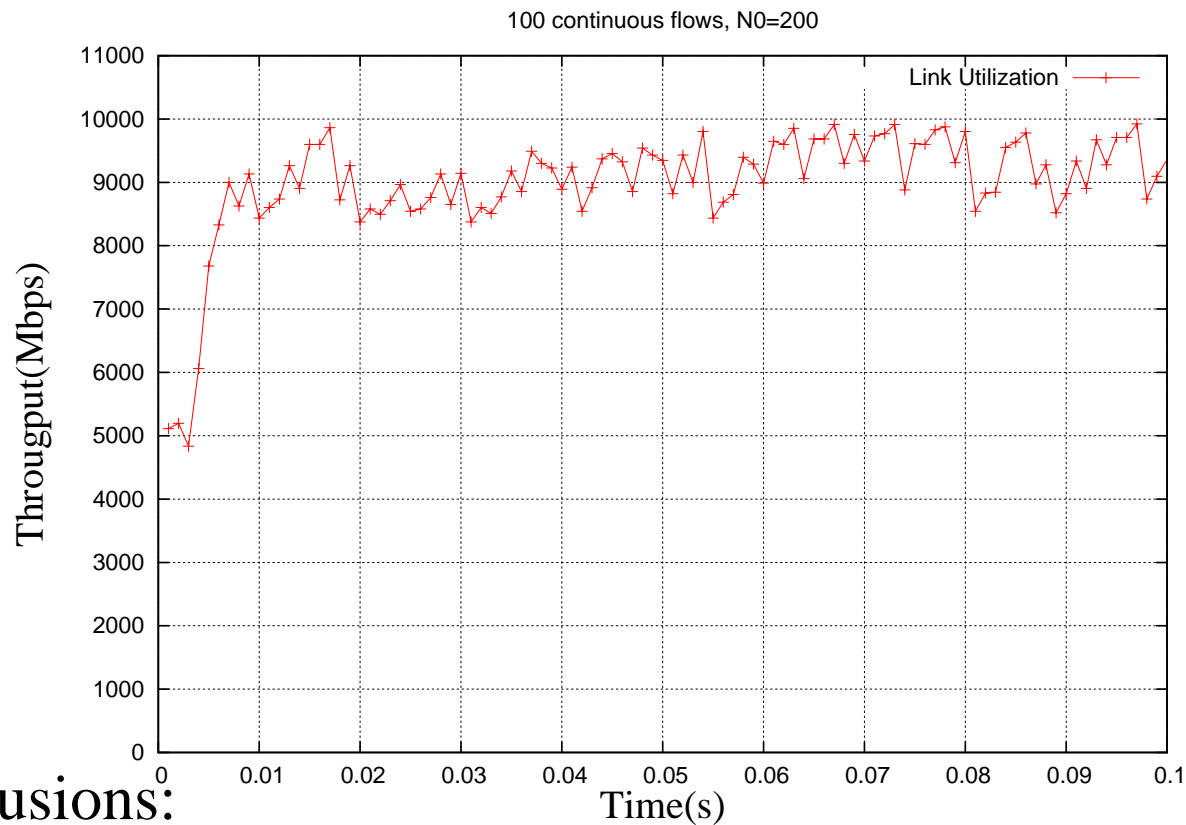


## Conclusions.

- Perfect Fairness!
- Fast Convergence: less than 10 ms



# Large Topology: Link Utilization

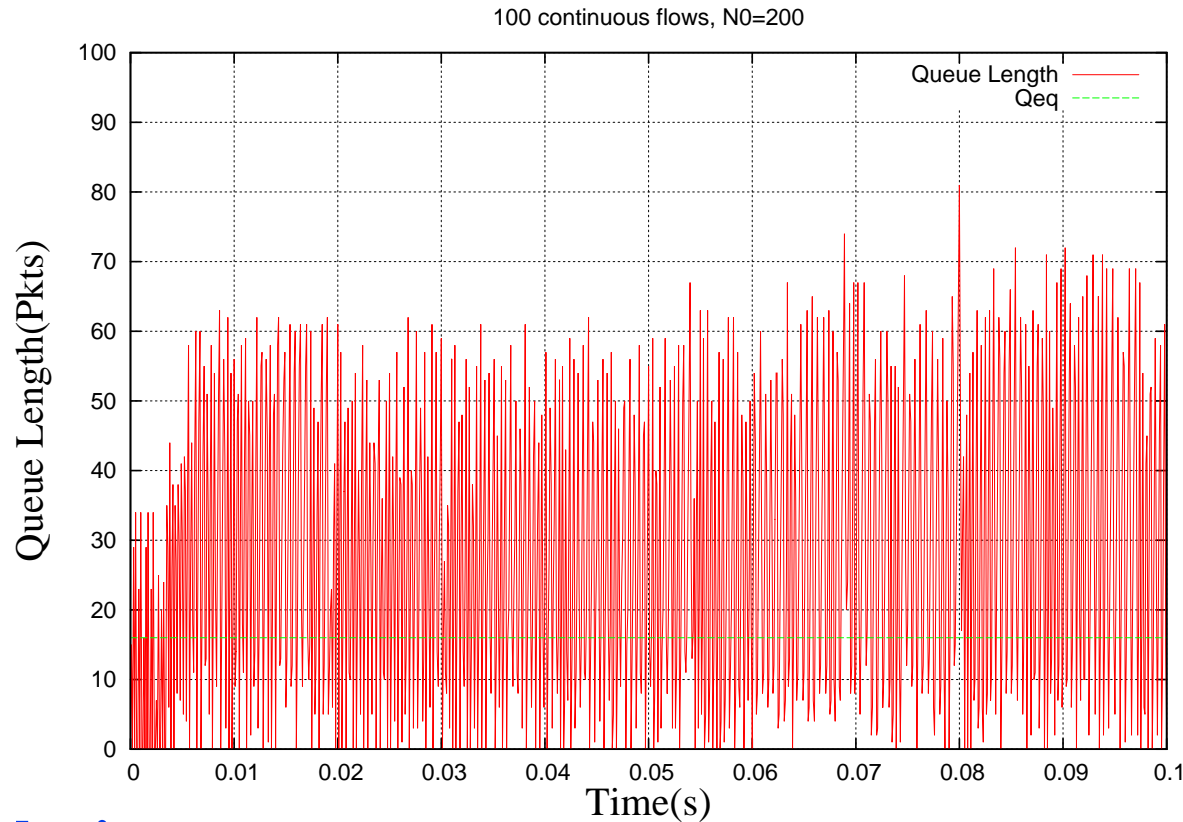


## Conclusions:

- The link is still 90+% utilized on average



# Large Topology: Queue Length

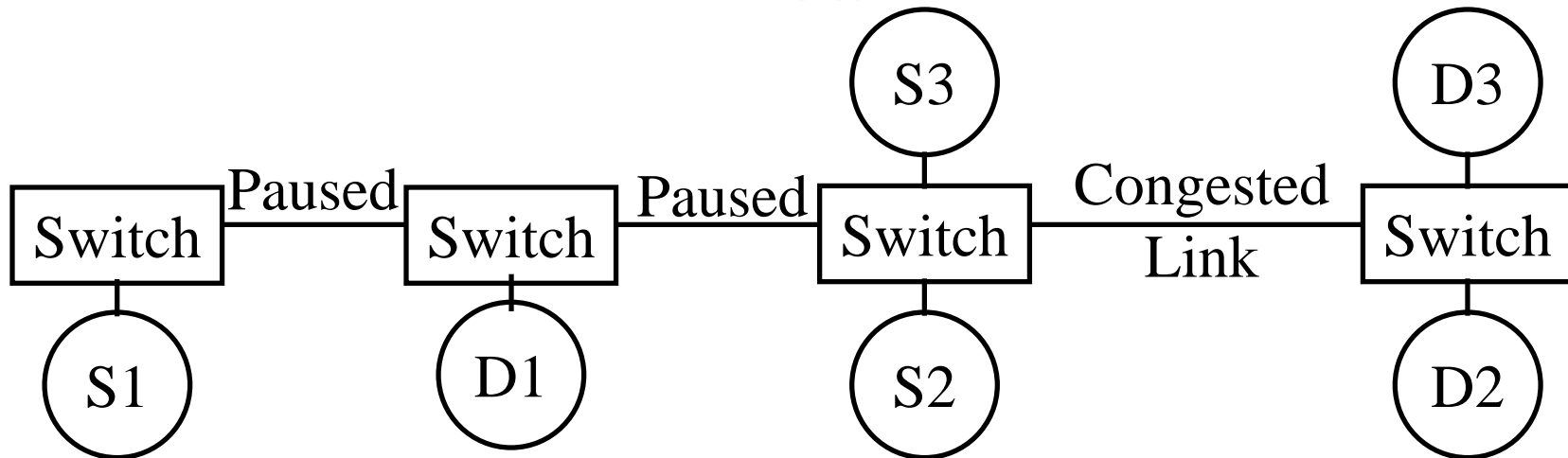


## □ Conclusions:

- Queue does not overflow!
- No PAUSE required or issued



# PAUSE

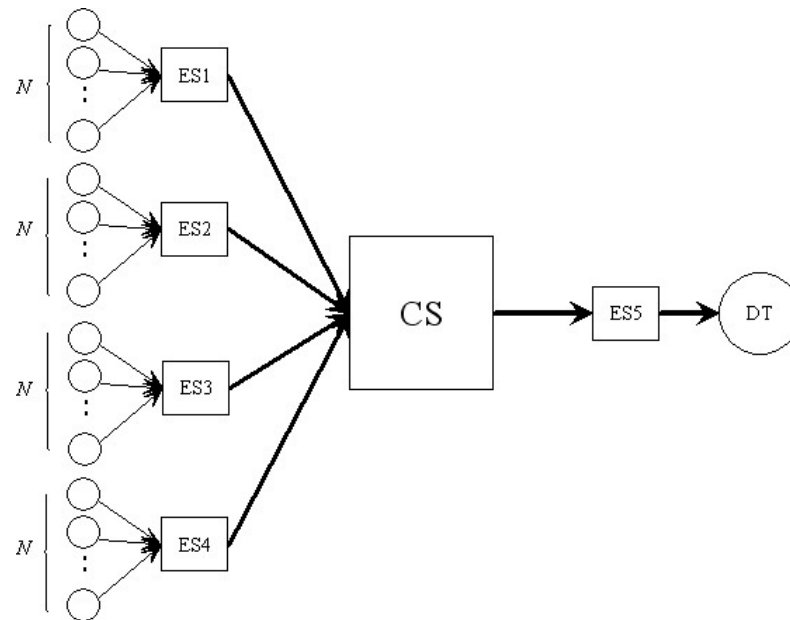


- ❑ S1-to-D1 flow is not using congested resources but is stopped by congestion caused by S2-to-D2 and S3-to-D3
- ❑ **Conclusion:**
  - ❑ Pause unfairly affects non-congestion causing flows
  - ❑ Pause should not be used as a primary or frequent mechanism
  - ❑ Pause can reduce loss but increase delays in the network
  - ❑ Pause is an emergency mechanism for rare use

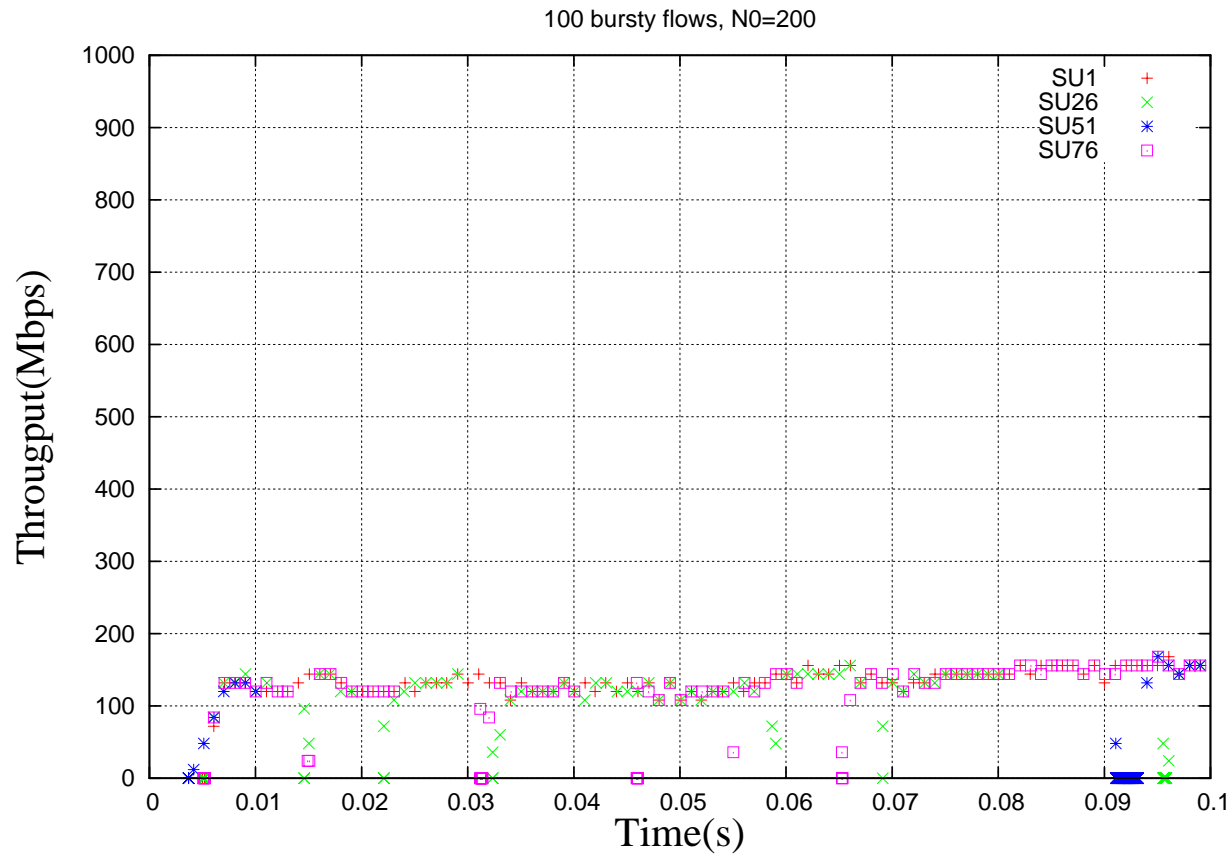


# Bursty Traffic: Configuration

- ❑ Large Topology (100 Sources)
- ❑ The sources come on and go off after transmitting a burst.
- ❑ The ON/OFF period is Pareto distributed
- ❑ Average ON/OFF period is *10* ms



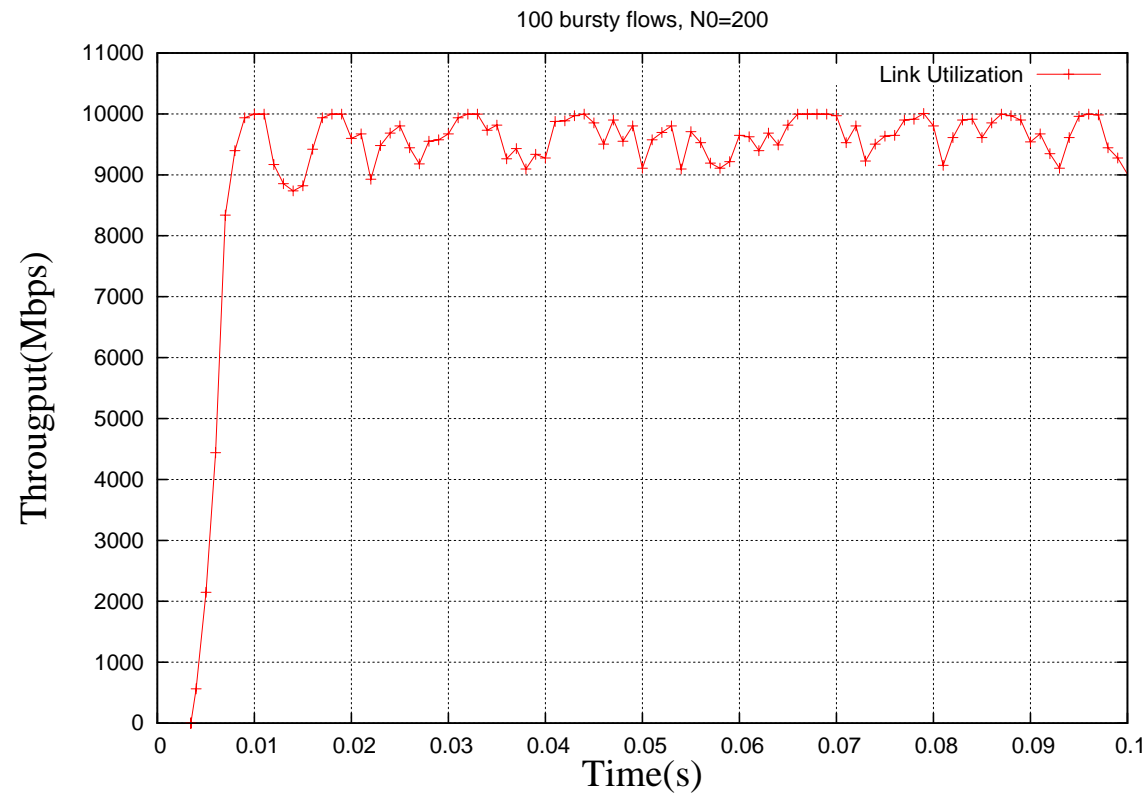
# Large Topology - Bursty Traffic: Throughput



□ **Conclusion:** Perfect Fairness!



# Large Topology - Bursty Traffic: Link Utilization

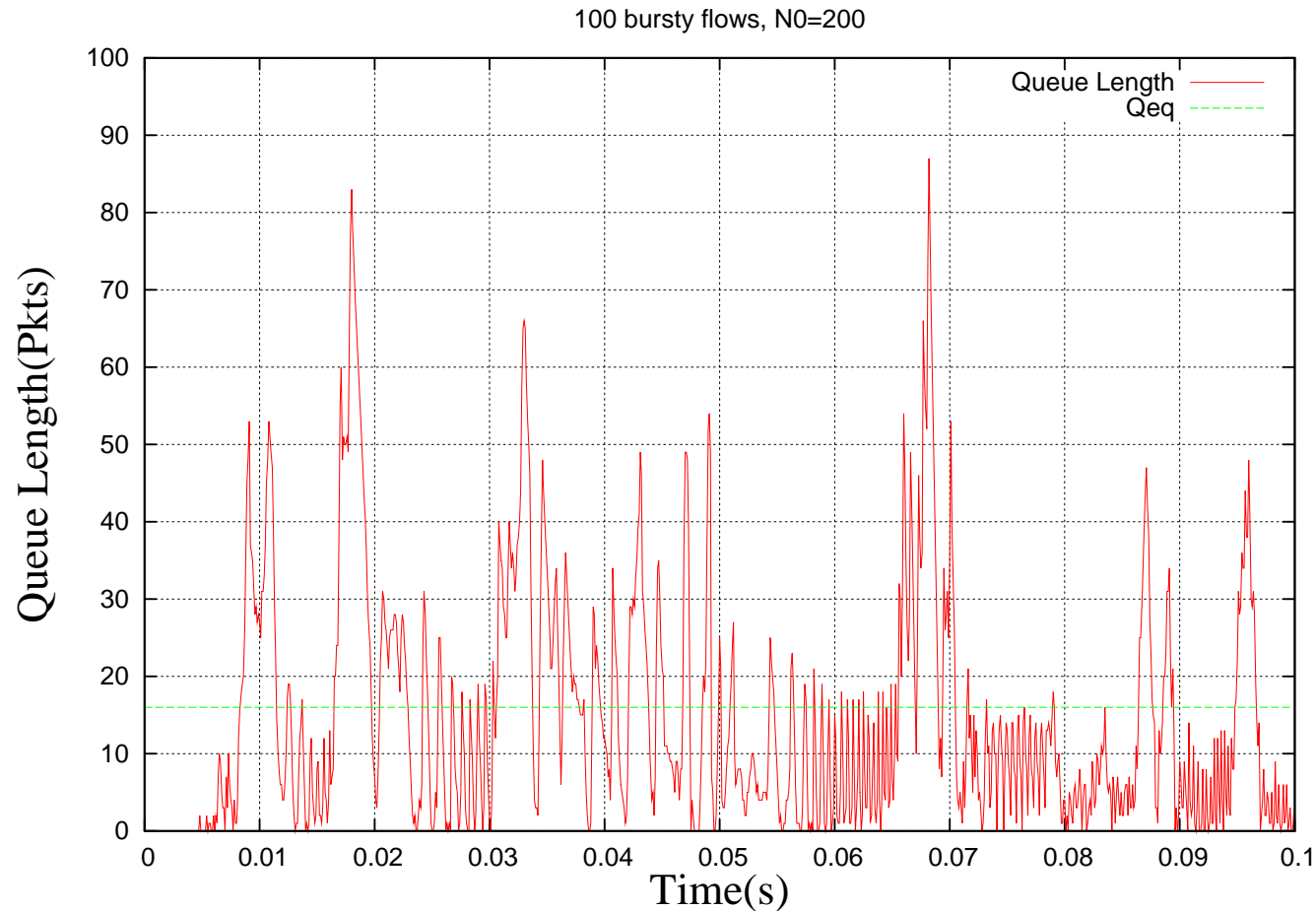


□ **Conclusion:** On average, the link is 95+% utilized





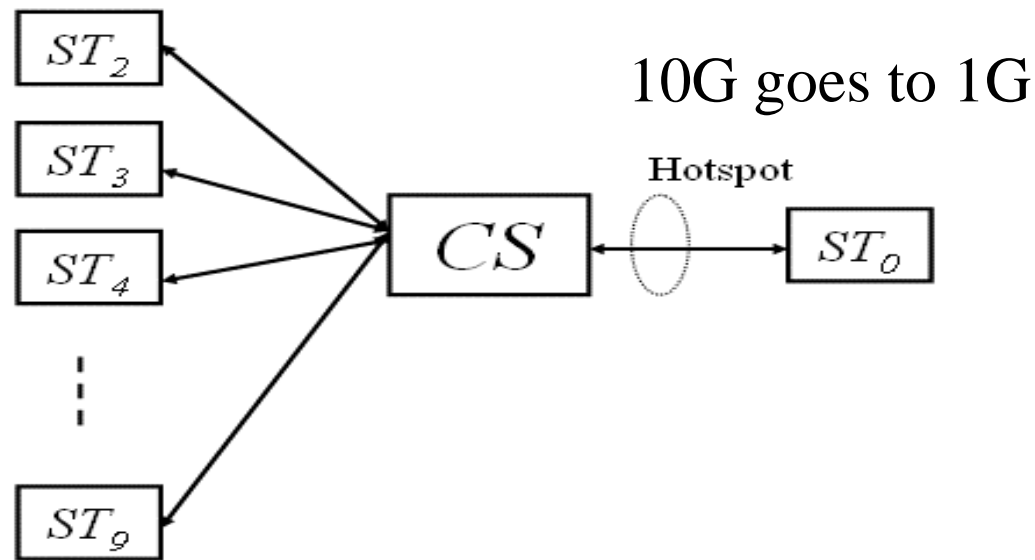
# Large Topology - Bursty Traffic: Queue



- ❑ **Conclusion:** Queue length is always under the buffer size.  
No pause required for this case.



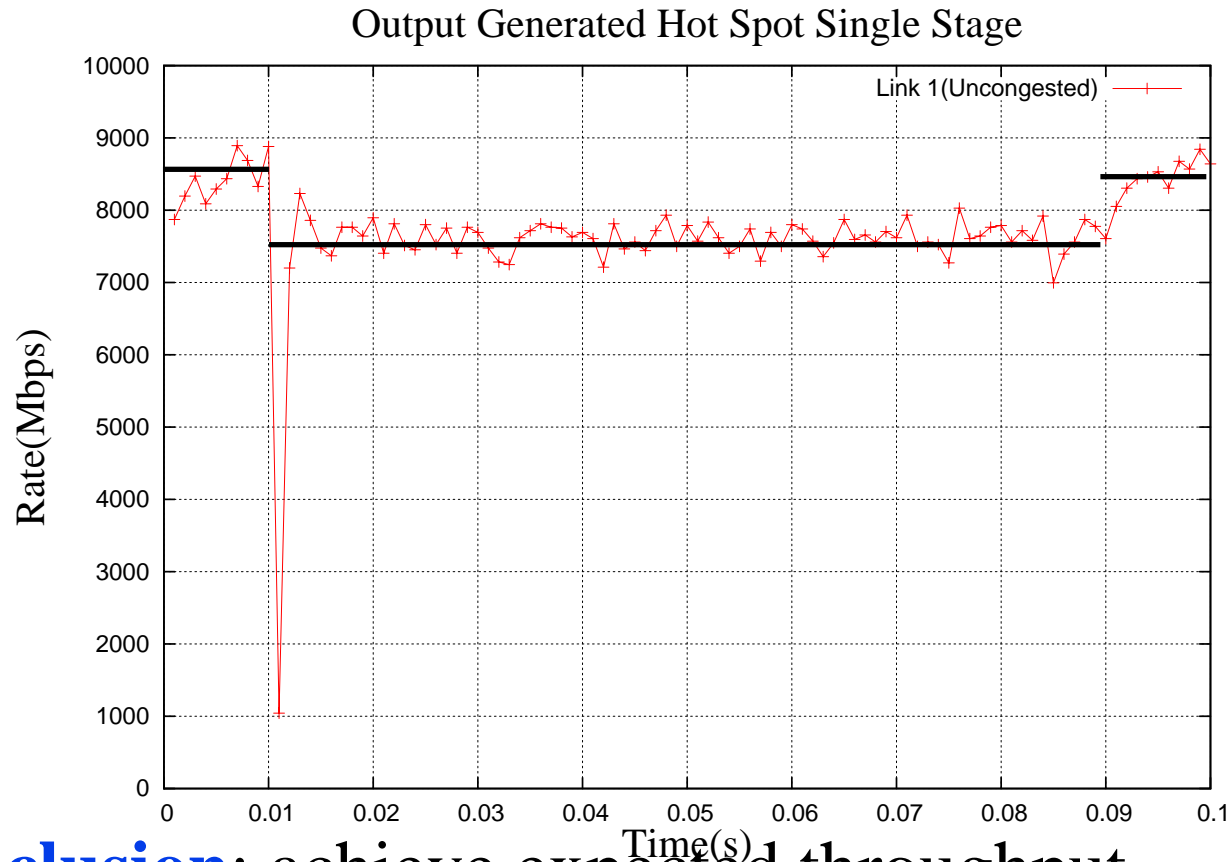
# Output Generated Hotspot Scenario



1. Capacity from  $CS$  to  $ST_0$  goes to **1** G from 0.01s to 0.09 s, then come back to *10* Gbps
2. We study per flow behavior instead of per node behavior
3. Symmetric topology configuration is used
4. Capacity  $C(t)$  is known from the idle time and bits transmitted.



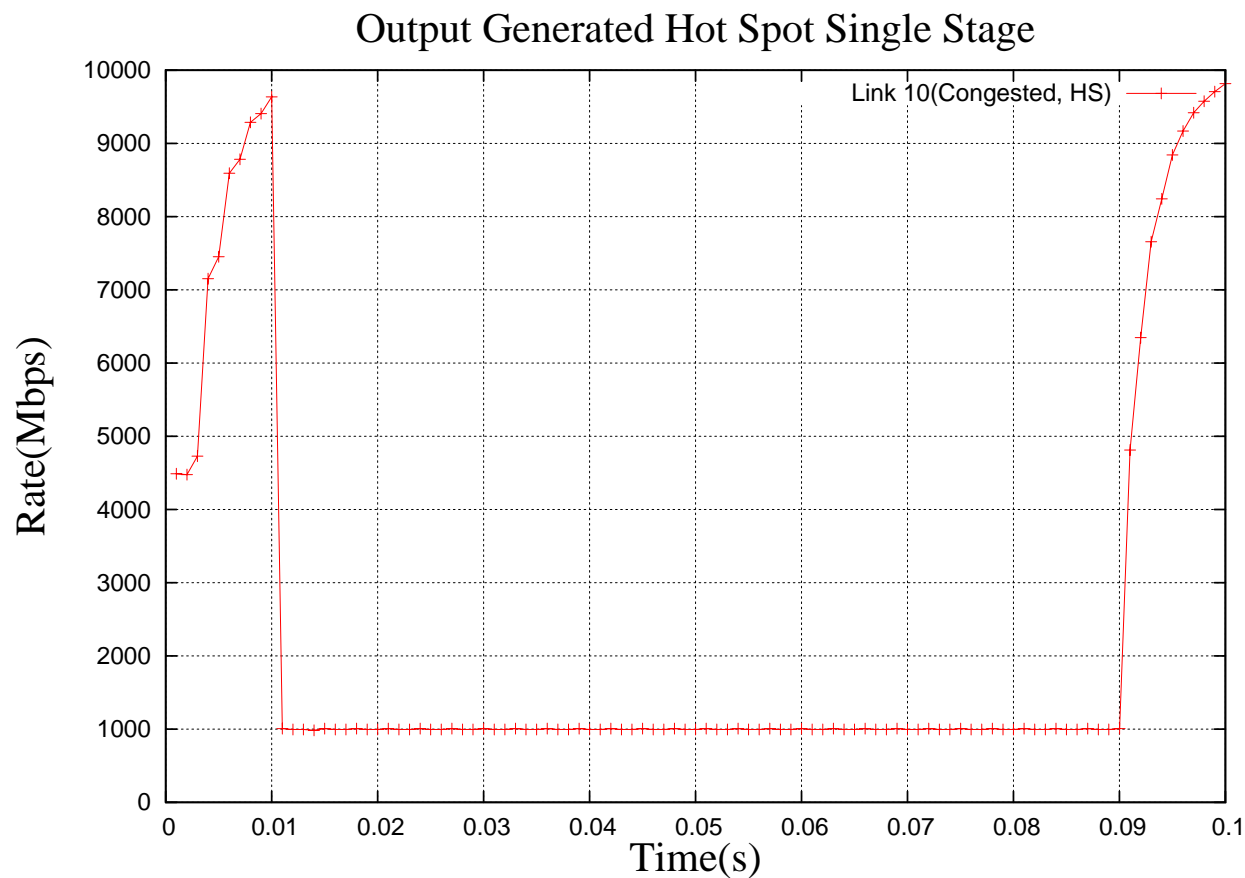
# OGHS – Uncongested Link (ST2 to CS)



□ **Conclusion:** achieve expected throughput



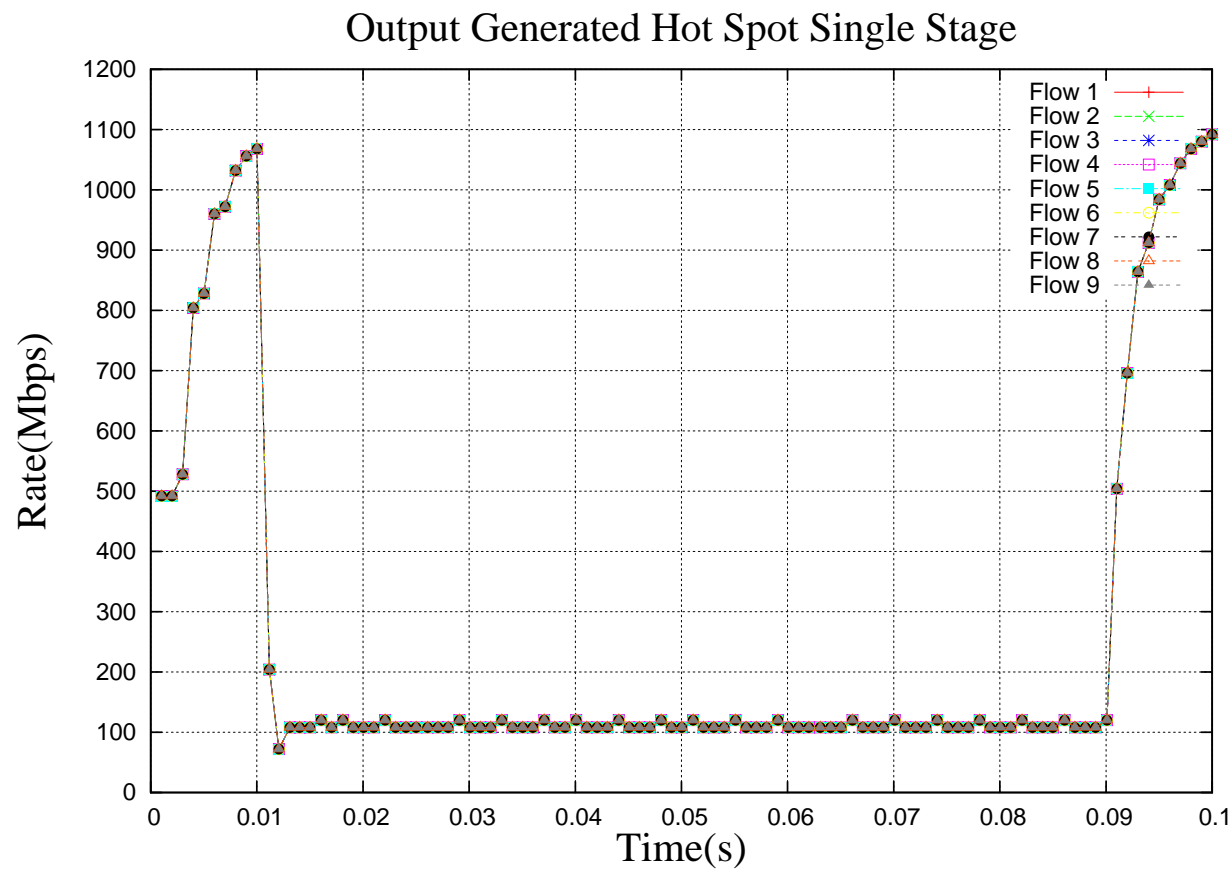
# OGHS – Congested Link (CS to ST0)



□ **Conclusion:** Fast convergence. Highly utilized link!



# OGHS – 9 Congested Flows

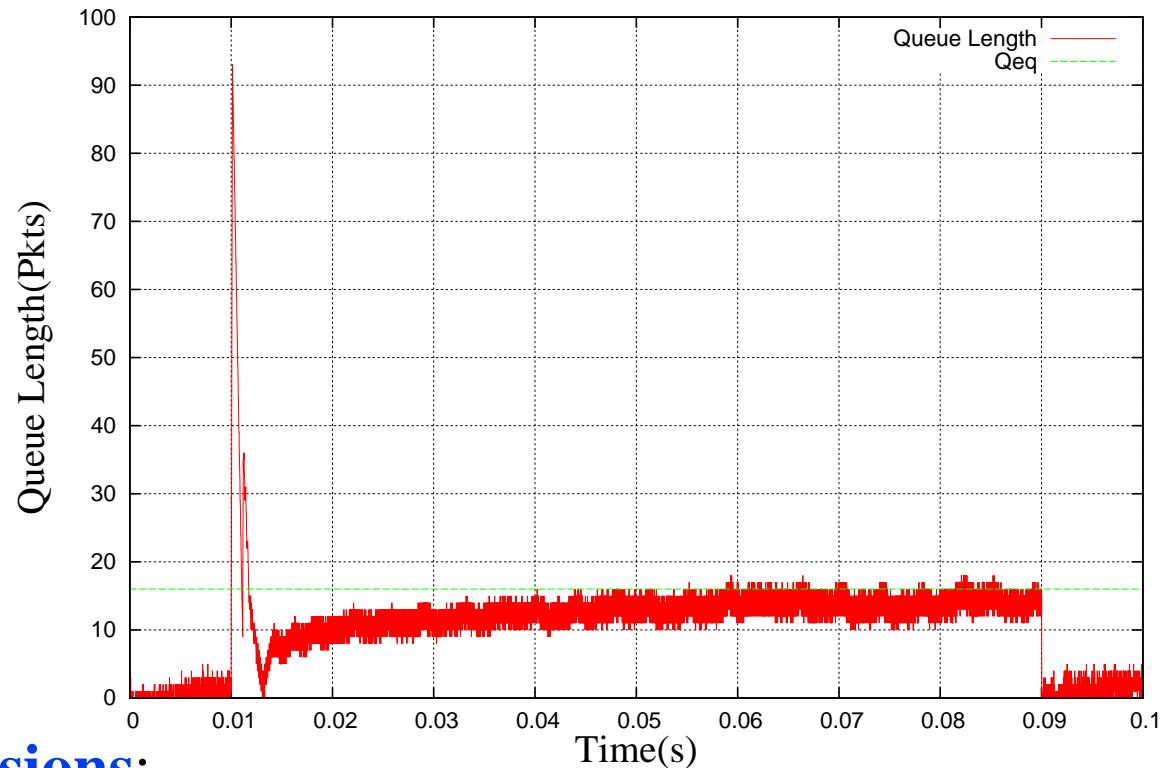


□ **Conclusion:** Perfect fairness among 9 flows!



# OGHS - Queue Length

Output Generated Hot Spot Single Stage

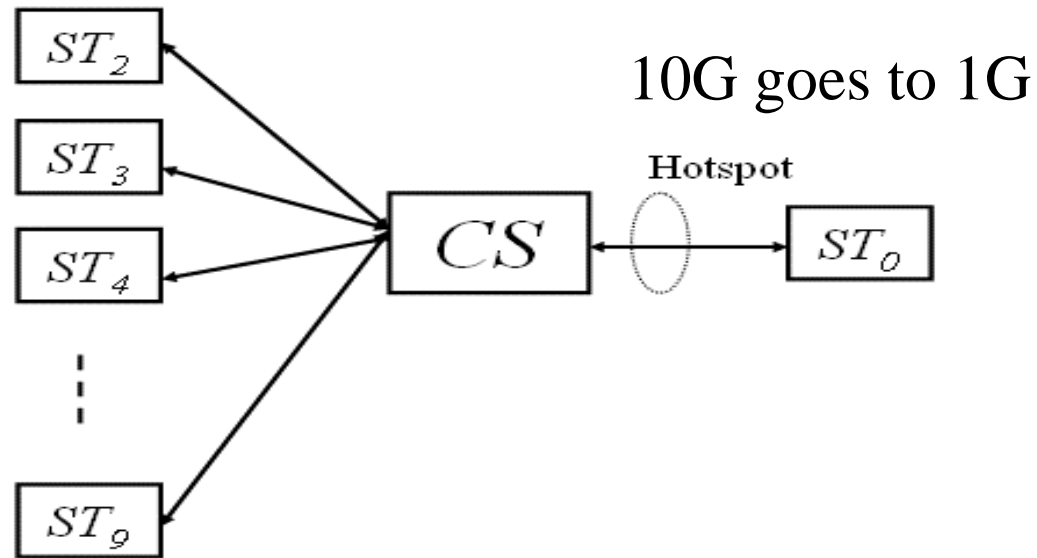


## Conclusions:

- ❑ One very short Pause event (capacity reduced from 10G to 1G).
- ❑ The queue is very stable at the equilibrium point



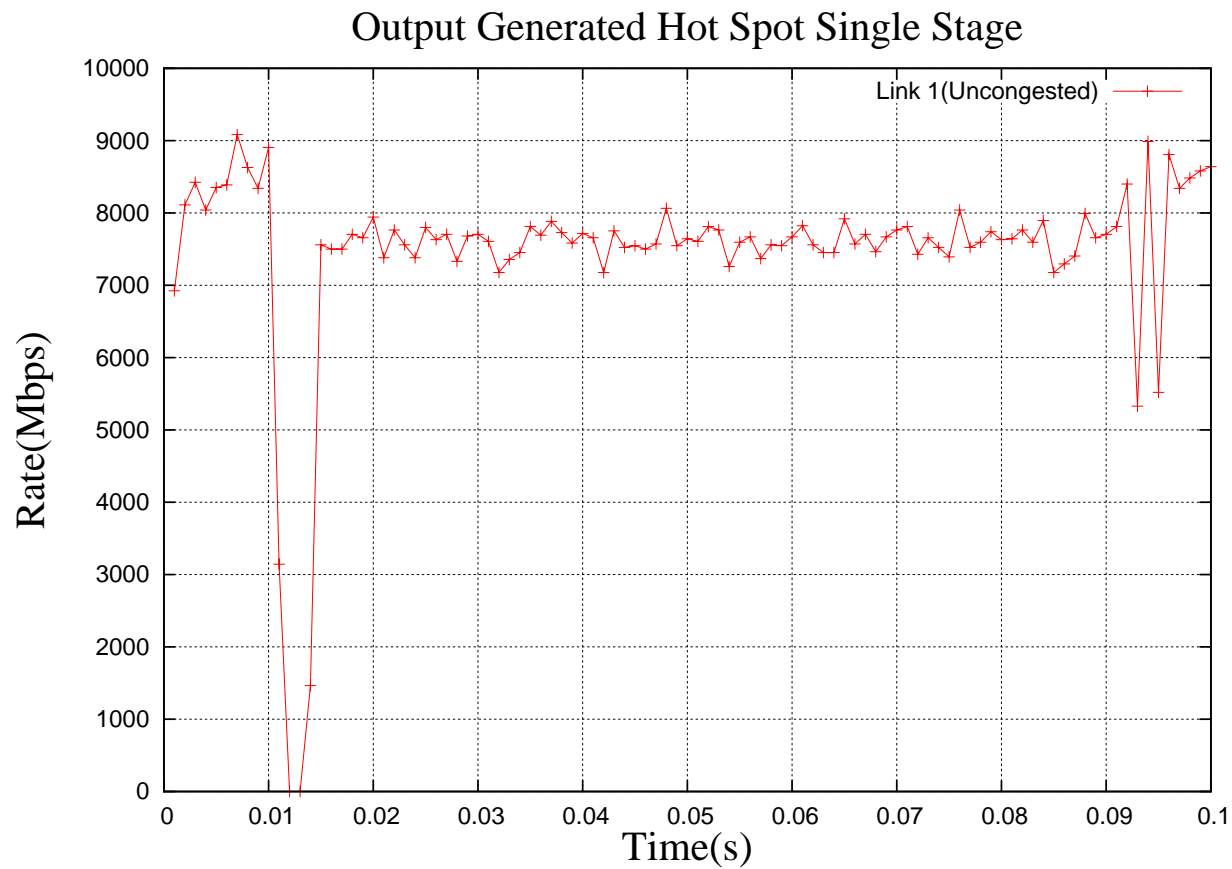
# OGHS – Long Delay



1. Each link ?? us long
2. Total feedback delay = 500 us



# OGHS-LD: Uncongested Link - ST2 to CS

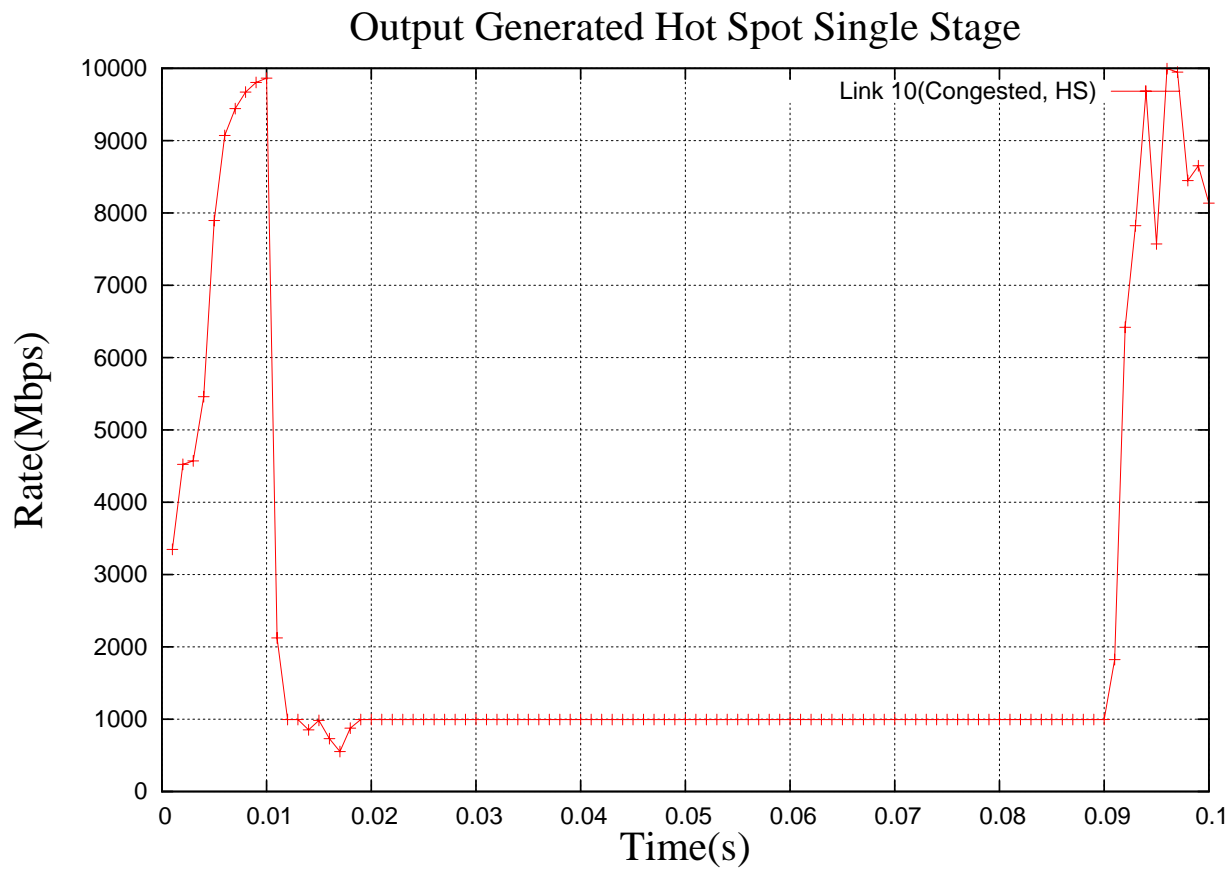


**Conclusion:** FECN recovers quickly





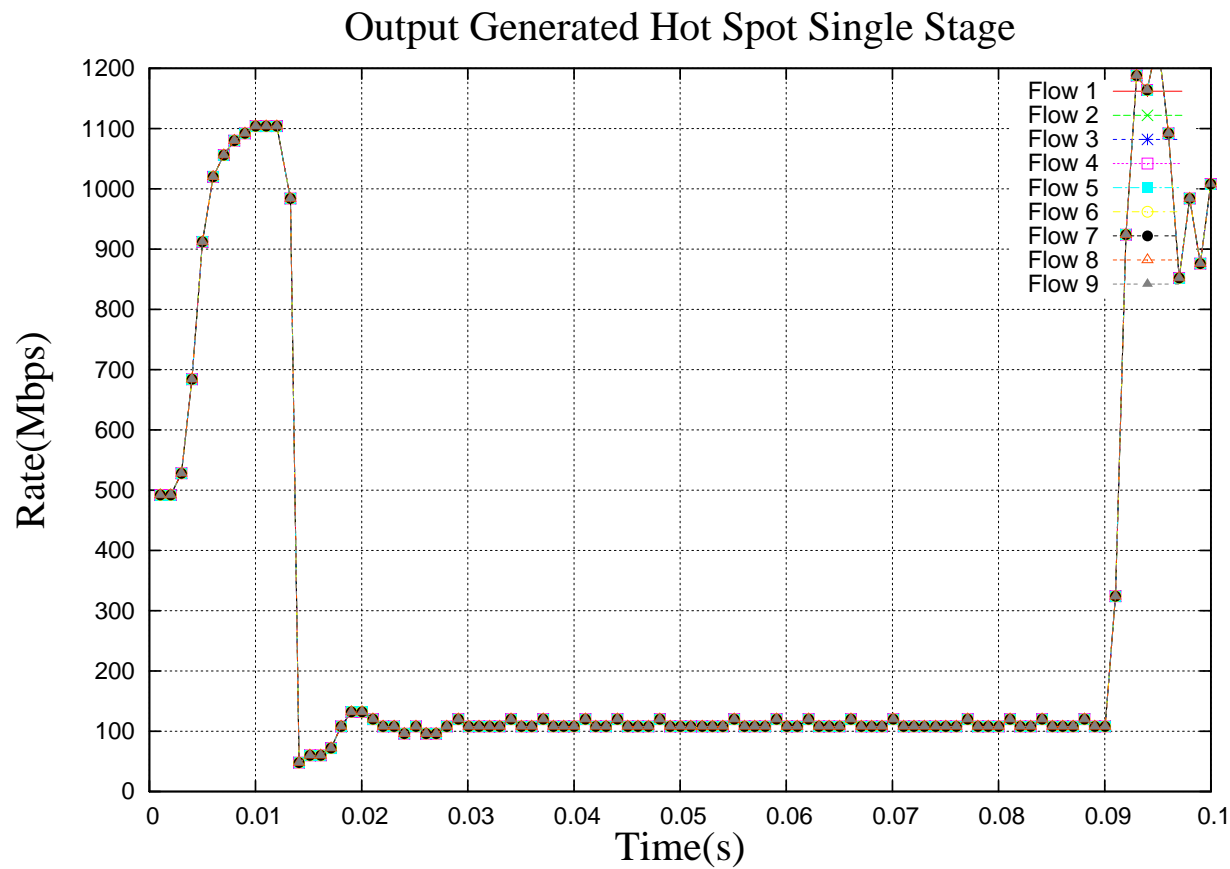
# OGHS-LD: Congested Link (CS to ST0)



□ **Conclusion:** Link throughput is stable.



# OGHS-LD: 9 Congested Flows

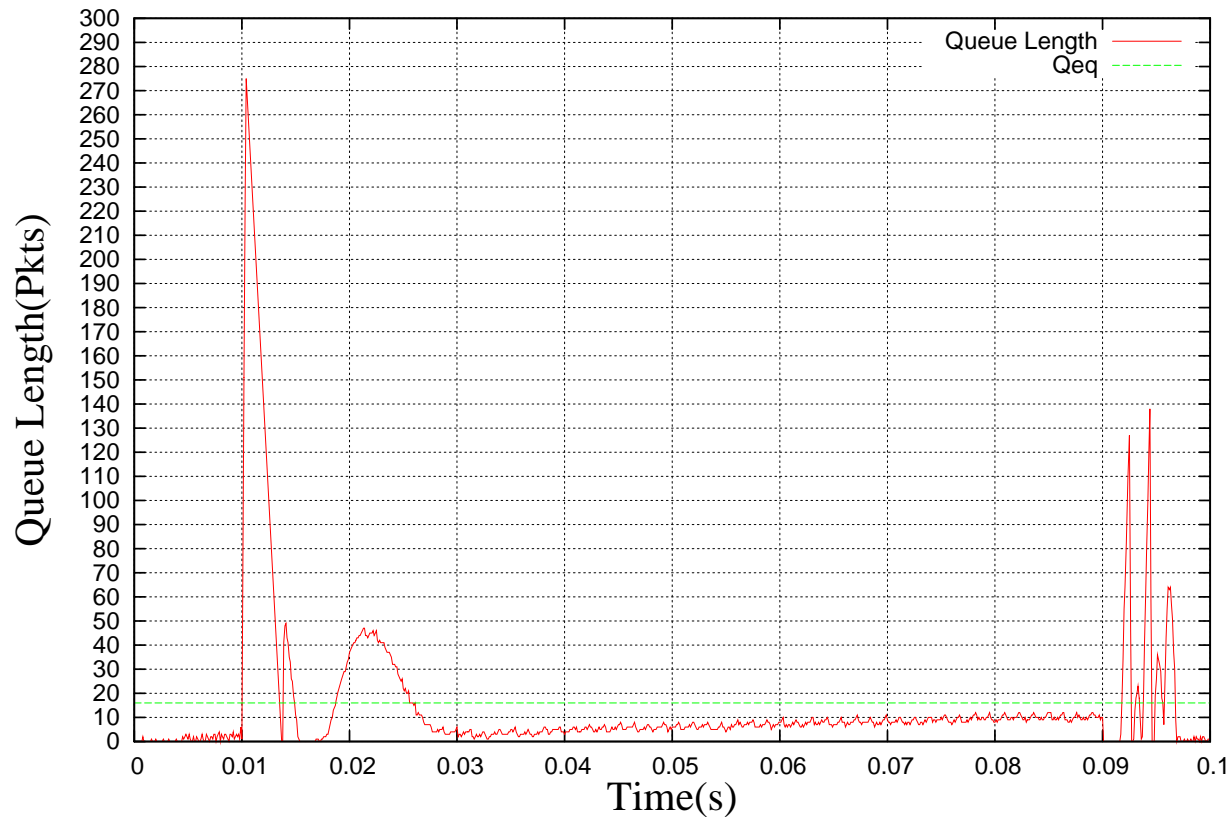


□ **Conclusion:** Perfect fairness



# OGHS-LD: Queue Length

Output Generated Hot Spot Single Stage



- ❑ **Conclusion:** FECN works for long delay without any change in any parameters



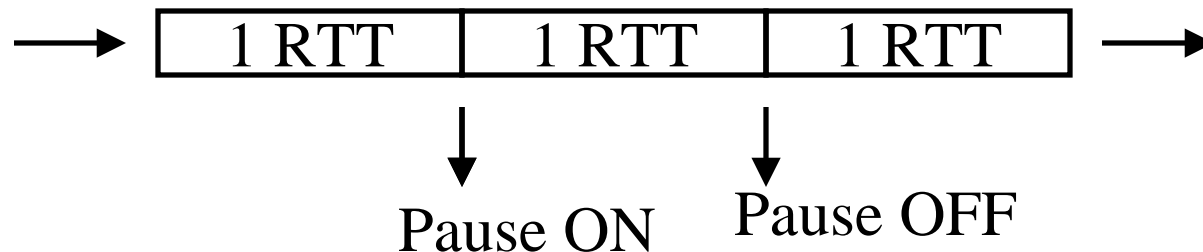
# OGHS-LD: Other Observations

- ❑ Pause On/Off Threshold is 90/8 packets...
- ❑ 3 Pause events
- ❑ Total pause duration 0.0045 s



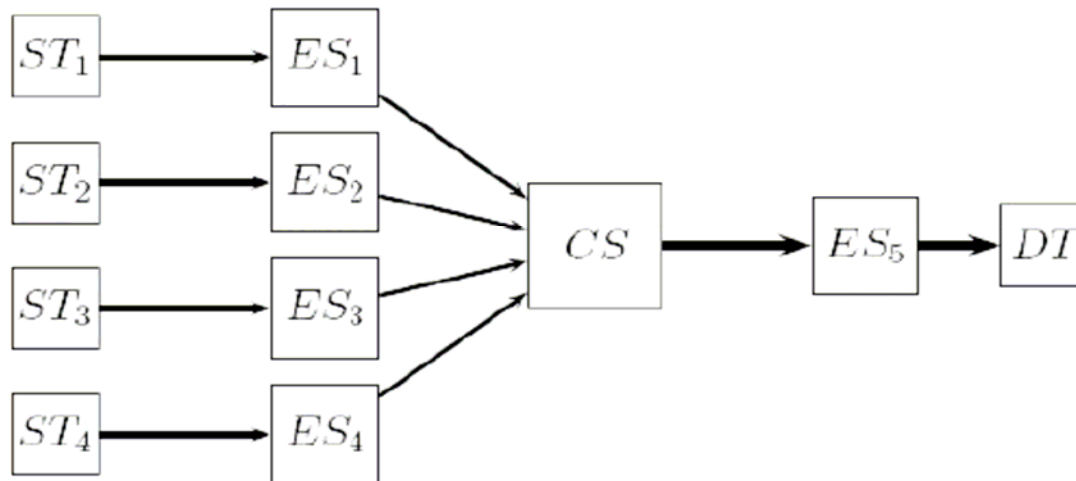
# Minimum Buffering Required w Pause

- ❑ Need 1 RTT buffer to allow queue to not go to zero after a Pause OFF
  - ⇒ Pause OFF threshold = 1 RTT
  - ⇒ Pause ON Threshold = 2 x Pause OFF = 2 RTT
- ❑ Need 1 RTT extra buffer to not drop any packets after a Pause ON
- ❑ Total Buffer = 3 RTT



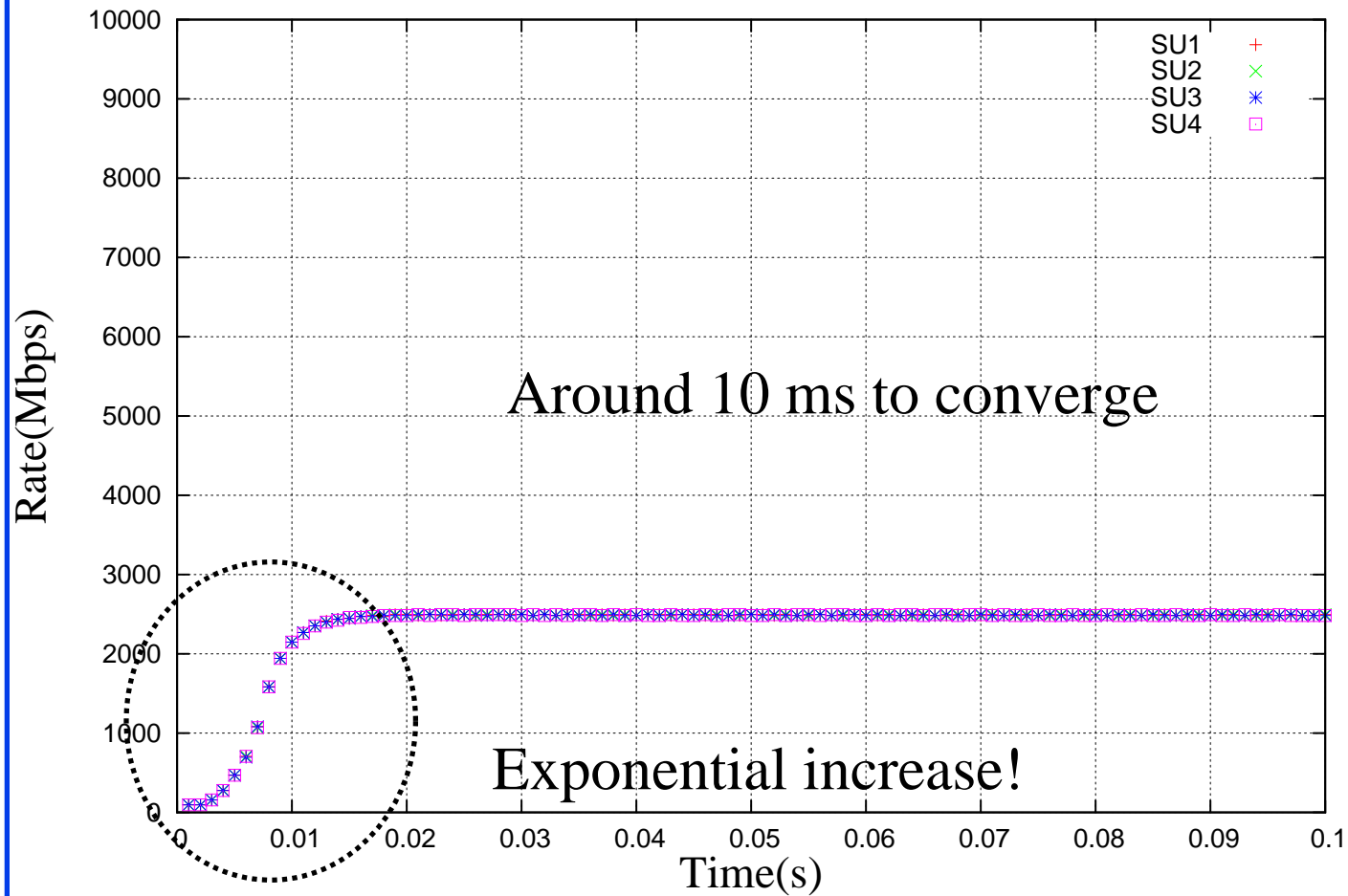
# Sensitivity Analysis

- ❑ All configurations analyzed so far used same parameter values, except for  $N_0$ .
- ❑ How  $N_0$  affects the scheme?
- ❑ Continuous traffic,  $N_0=100, 80, 40, 20$



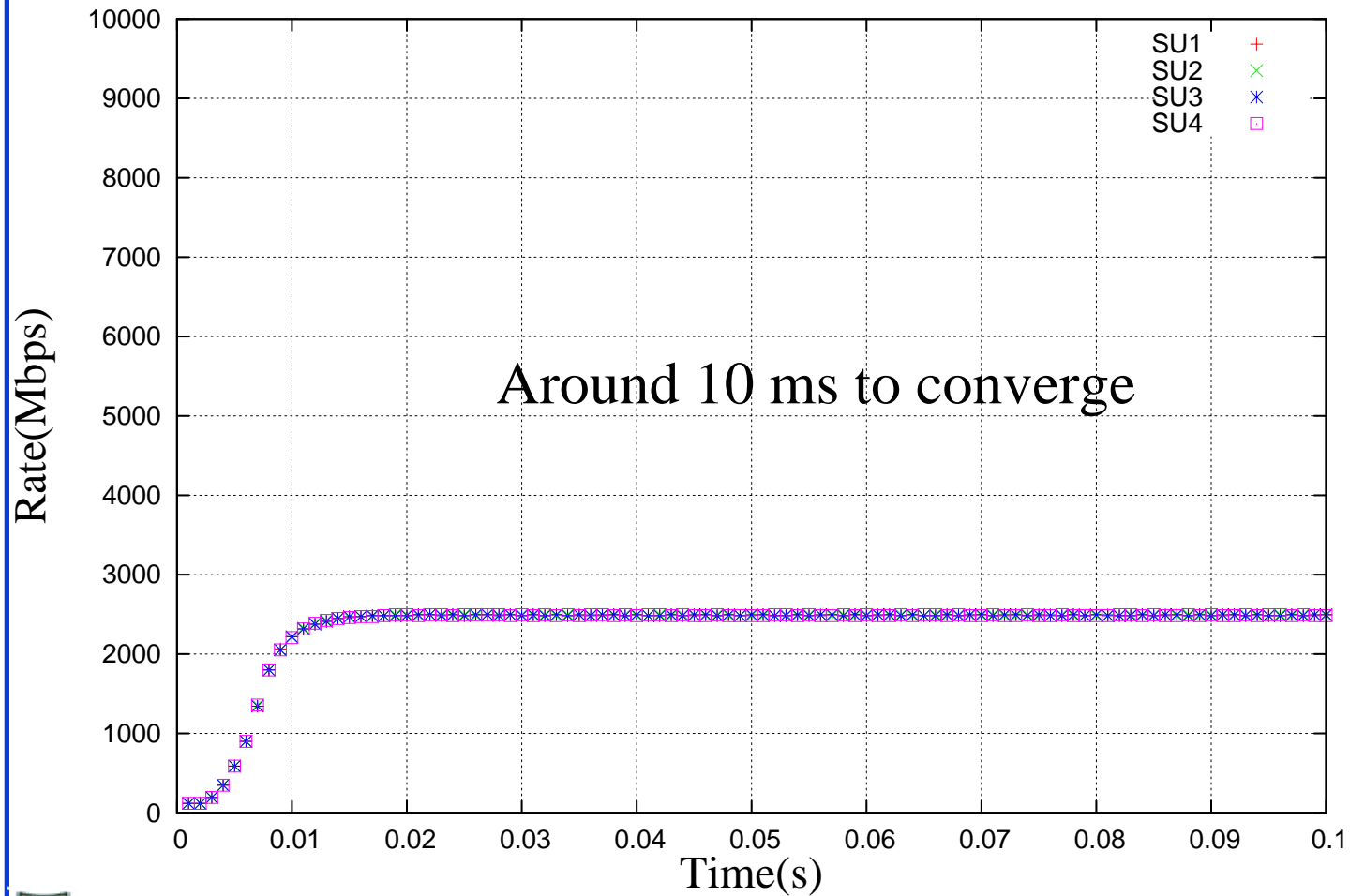
# N0=100

4 continuous flows, N0=100, alpha\_1=0.5, alpha\_2=0.5



# N0=80

4 continuous flows, N0=80, alpha\_1=0.5, alpha\_2=0.5



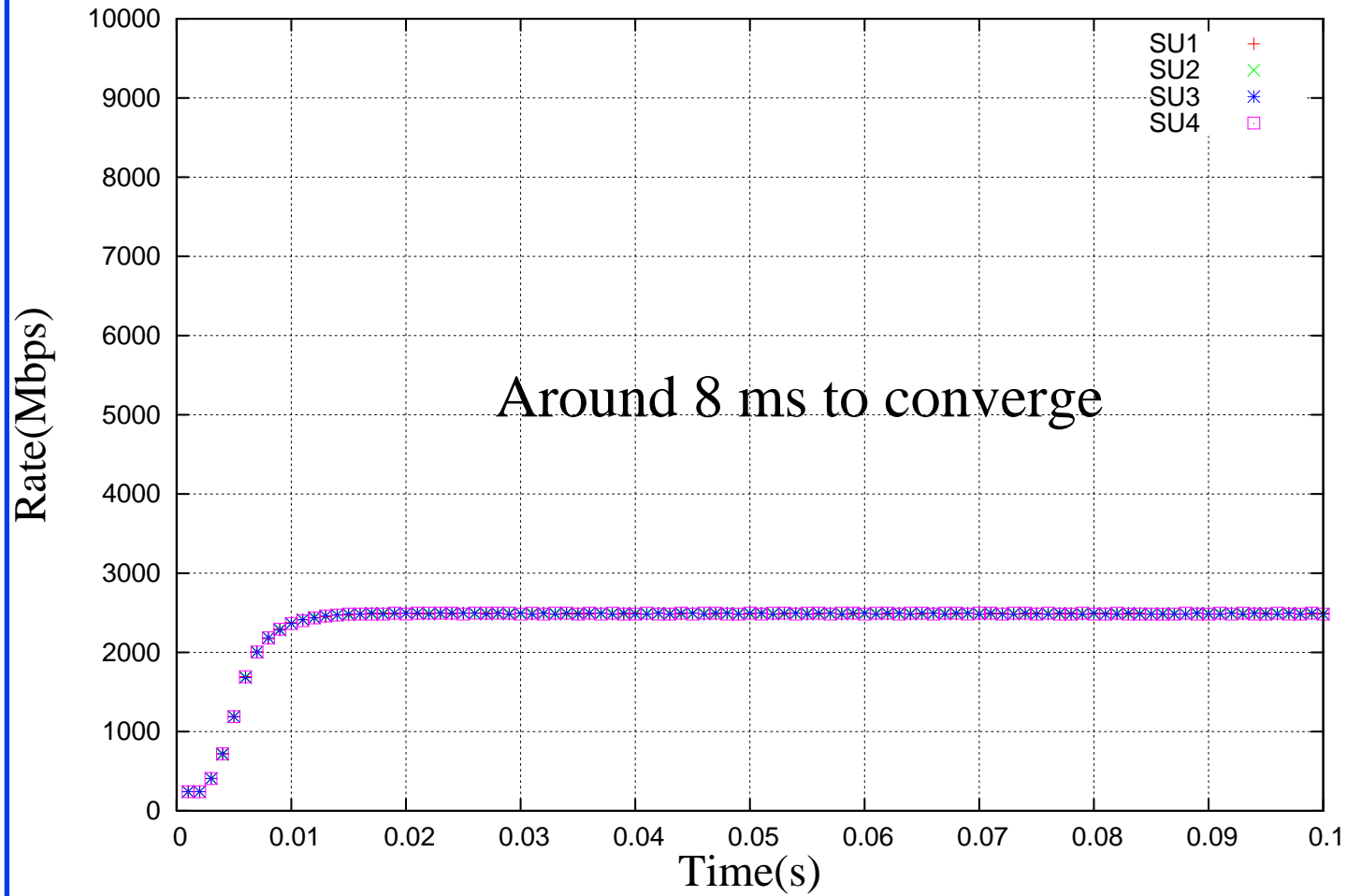
Around 10 ms to converge





# N0=40

4 continuous flows, N0=40, alpha\_1=0.5, alpha\_2=0.5

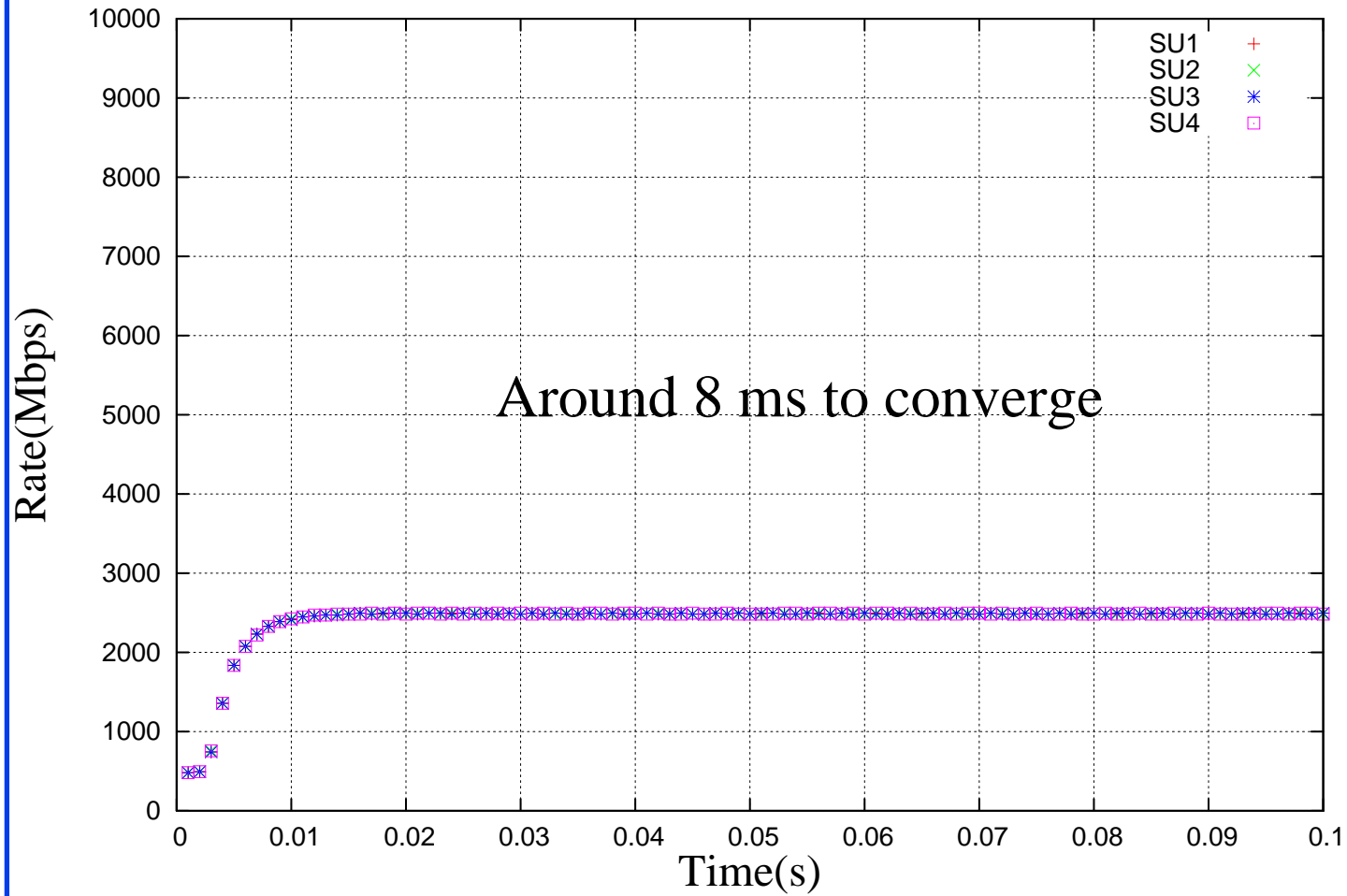


Around 8 ms to converge



# N0=20

4 continuous flows, N0=20, alpha\_1=0.5, alpha\_2=0.5



Around 8 ms to converge



## Sensitivity to N0

- The limited rate increase results in a logarithmic rise

Convergence time to go from N0 to N =  $\log_{\lambda} \left( \frac{N0}{N} \right)$

Here  $\lambda$  is the multiplier used in limited increase

So now N0 does not have a significant effect. It can be set to a large value.



# Overhead of FECN

- ❑ Given the configuration of the network, FECN has almost deterministic overhead
- ❑ Each flow generates one tag every  $T$  interval.
- ❑ For  $N$  flows in a simulation of duration  $t$ :
  - ❑  $t*N/T$  FECN tags are added to forward data packets
  - ❑  $t*N/T$  FECN control messages returned by the destinations
- ❑ Alternative designs where  $T$  is dynamically varied depending upon the stability, load, or rate were tried successfully but deemed unnecessary. A simple two  $T$  strategy consists of using a larger  $T$  if the system is operating near optimal region.
- ❑ It is also possible to use count based rate discovery, where every  $n$ th packet is tagged. This works but convergence to fairness takes slightly longer.

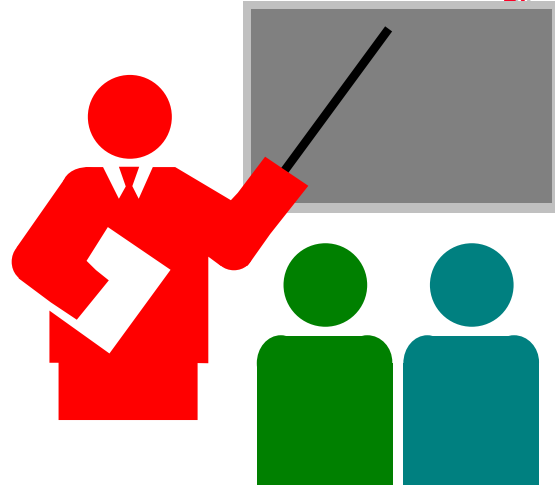


# Advantages of FECN

- ❑ Flexibility:
  - ❑ Switches can base rates on resources other than one queue, e.g., sum of input and output queues, utilization of shared buffers, # of channels available on a wireless link, etc.
  - ❑ Switches can give different rate to a flow based on traffic type, class of service, types of sources, VLANs
- ❑ Works perfectly on variable link speeds, e.g., wireless links
- ❑ Vendor differentiation



# Summary



1. Convergence of rates is very fast
2. Convergence time is a small multiple of measurement interval  $T$
3. Convergence to fairness is built in. All active sources get the same rate.
4. **Bursty traffic** can be supported and can get fair and efficient allocation due to fast convergence



## Summary (Cont)

5. RD tags in the packets are simple – just rates, RLQ ID, and direction.
6. Source algorithm is quite simple
7. Switch enhancements minimize queue buildup and avoid the need for PAUSE
8. No internal parameters or details of the switch are shared outside with the sources  $\Rightarrow$  Switch algorithms and parameters can be easily changed
9. Very few parameters: T
10. Parameters are easy to set.
11. Scheme not very sensitive to parameters
12. Potential for vendor differentiation for switch algorithms.



# References

- Bobby Vandalore, Raj Jain, Rohit Goyal, Sonia Fahmy, "Dynamic Queue Control Functions for ATM ABR Switch Schemes: Design and Analysis," Computer Networks, August 1999, Vol. 31, Issue 18, pp. 1935-1949.  
[http://www.cse.wustl.edu/~jain/papers/cnis\\_qctrl.htm](http://www.cse.wustl.edu/~jain/papers/cnis_qctrl.htm)

