



# Evaluation of TCP congestion control mechanisms using OPNET simulator

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ENSC 835: COMMUNICATION NETWORKS

FINAL PROJECT PRESENTATION  
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# Roadmap

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- Introduction
- Objective and scope
- Implementation
- Simulation
- Conclusion
- Future work
- References



# Introduction: Transmission Control Protocol

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- reliable, error free, connection-oriented, point-to-point with flow control
- used by most applications: FTP, HTTP, SMTP
- carries about 90% internet traffic [1]
- developed based on wired network properties
- Sliding window protocol for flow control
- problems exist when an intermediate router run out of memory and drops the packets termed as **Congestion**
- no information from lower layers on congestion

[1] I. Khalifa and Lj. Trajkovic, "An overview and comparison of analytical TCP models," (invited session) *Proc. IEEE Int. Symp. Circuits and Systems*, Vancouver, British Columbia, Canada, May 2004, vol. V, pp. 469-472.



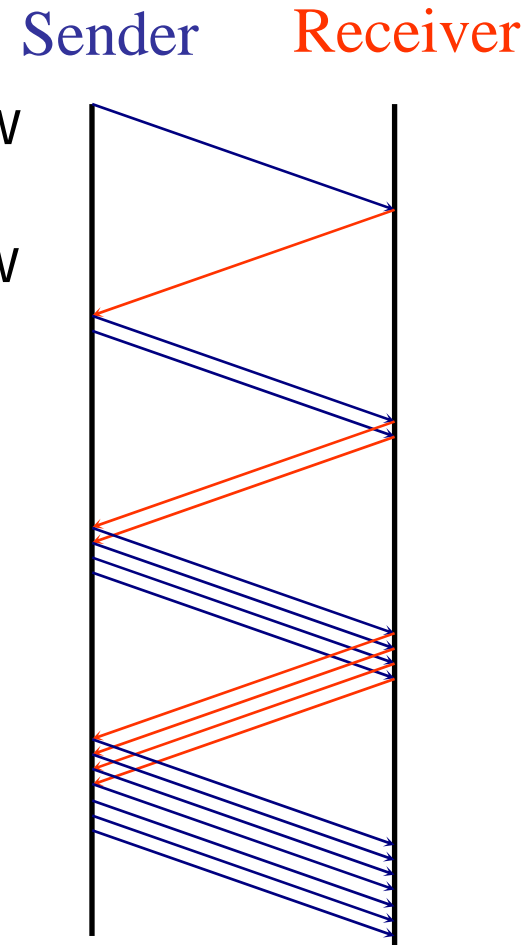
# Introduction: Transmission Control Protocol

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- packet loss and delay indicate congestion
- TCP maintains congestion window accordingly
- Congestion window:
  - maximum number of bytes transmitted without ACKs sent packets
  - minimize router buffer overflow and retransmission effort
- Approaches to manage congestion window:
  - Slow Start
  - Additive Increase Multiplicative Decrease
  - Congestion Avoidance
  - Fast retransmit and Fast recovery

# Introduction: Slow Start

- initial congestion window: 1 MSS
- on receiving ACK, congestion window is set to 2
- process increases congestion window exponentially
- continues exponential increase until loss event or advertised receiver window, whichever is minimum
- rapid utilization of available bandwidth.





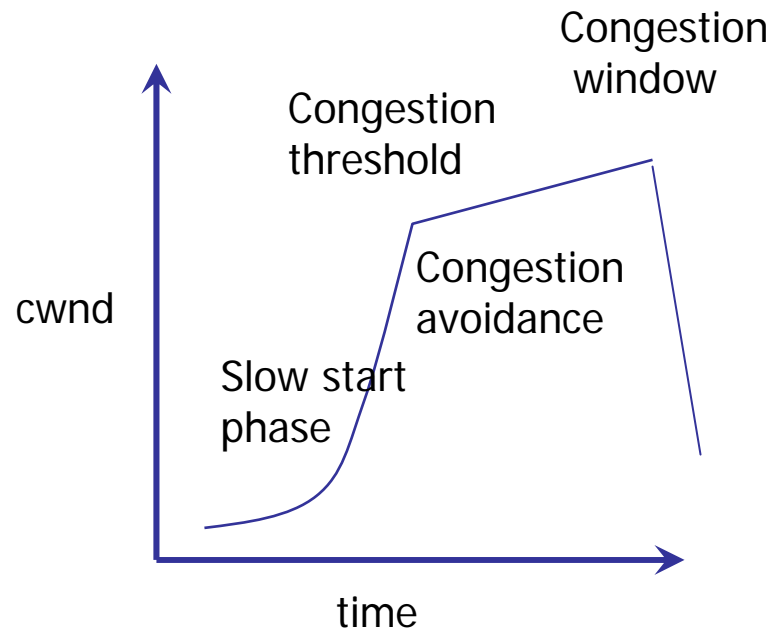
# Introduction: AIMD

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- Additive Increase:
  - under no congestion, window increases by 1 MSS for every RTT (Rather than for every ACK)
  - helps in additional use of bandwidth
- Multiplicative Decrease:
  - on loss event, window reduced to half of the current window size
  - continues to decrease half of the previous window on successive drops
  - minimum window size is 1 MSS

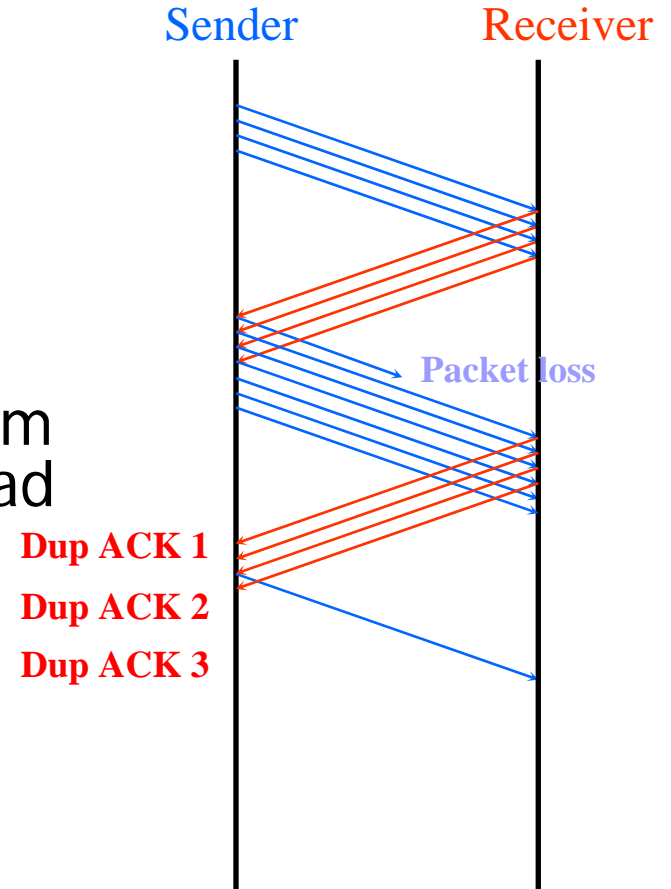
# Introduction: Congestion Avoidance

- after slow start threshold, window increases by 1 MSS for every RTT rather than exponentially



# Introduction: Fast retransmit and Fast recovery

- Fast retransmit:
  - uses duplicate Acks to retransmit
  - retransmits without waiting for timeout
- Fast recovery:
  - after fast retransmit, perform congestion avoidance instead of slow start
  - duplicate ACK indicates availability of network resources







# Introduction: Algorithms

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- Reno, New-Reno, and SACK
  - use slow start, congestion avoidance, fast retransmission and fast recovery
- Reno
  - remains in fast recovery until it receives duplicate acknowledgments
  - transmits new packet for additional received duplicate acknowledgment
- SACK
  - acknowledge out-of-order segment selectively rather than cumulatively



# Introduction: Algorithms

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- NewReno
  - improves retransmission during fast recovery
  - does not exit fast-recovery until acknowledgment of all data which are not acknowledged while entering fast recovery
  - uses partial ACKs (ACK segments that do not acknowledge all the information that has been sent up to the moment when they are issued)



# Introduction: Literature Review

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- “New-Reno TCP in absence of SACK avoids Reno TCP's performance problems when multiple packets are dropped from a window of data. But still SACK is better in case of multiple packet drops.” [2]
- “SACK performance deteriorates in case of congested network. In case of non-congested network all three algorithms have comparable performance.” [3]

[2] S. Floyd and K. Fall, “Simulation Based Comparisons of Tahoe, Reno and Sack TCP”, ACM Computer Communication Review, 1996, Vol.26, No.3: 5–21.

[3] R. Paul and Lj. Trajkovic, "Selective-TCP for wired/wireless networks," Proc. SPECTS 2006, Calgary, AL, Canada, Aug. 2006, pp. 339-346.



# Introduction: Literature Review

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- New-Reno outperforms Reno and SACK when no packet losses occur during the slow-start phase. Bandwidth-delay product leads to performance degradation regardless of TCP versions and the bottleneck buffer size. [4]
- Performance of New-Reno is worse than Tahoe. SACK is the best and the most robust over the wireless channel. [5]

[4] H. Lee, S. Lee and, Y. Choi, "The influence of the large bandwidth-delay product on TCP Reno, NewReno, and SACK", *Proc. Information Networking Conference*, Oita, Japan, 2001, pp. 327–334.

[5] F. Anjum and L. Tassiulas, "Comparative study of various TCP versions over a wireless link with correlated losses", *IEEE/ACM Transactions on Networking (TON)*, NJ, USA, June 2003, Vol. 11, Issue 3, pp. 370 – 383.



# Project: Objective and scope

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- Objective:
  - observe, analyze, and compare congestion window recovery processes
  - analyze congestion window in case of link disconnection (wireless property)
  - get familiar with OPNET simulation tool
- Scope:
  - compare Reno, SACK, and New-Reno
  - analyze Congestion window
  - simulate drop and disconnection events



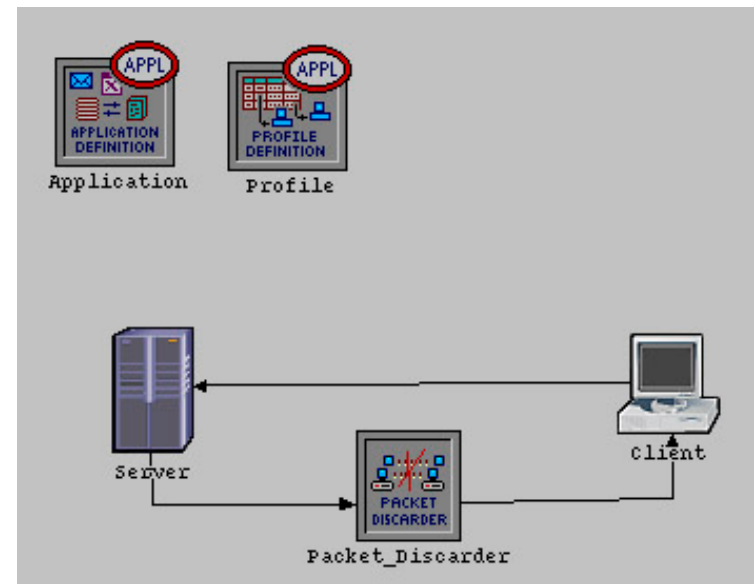
# Implementation: Steps

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- get familiar with Opnet tool
- get familiar with algorithms implemented
- select and built appropriate nodes
- built suitable network topology
- select appropriate parameter to evaluate performance of algorithms during congestion
- select appropriate statistics
- perform analysis on collected statistics

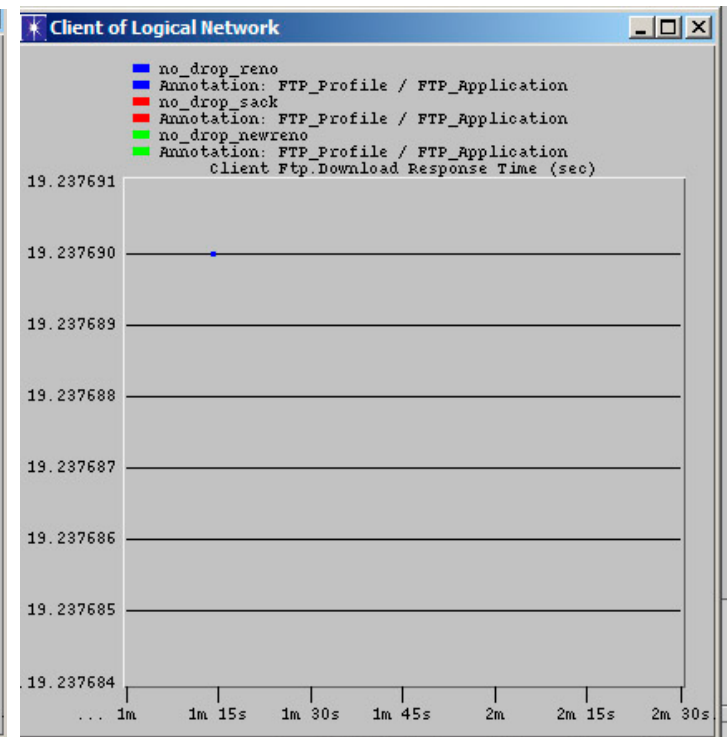
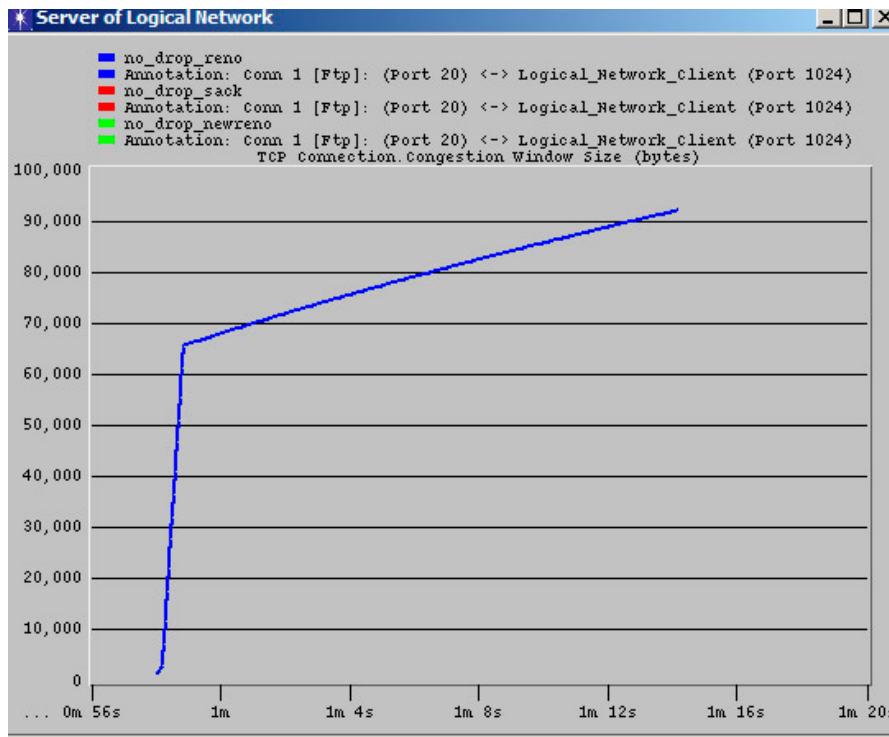
# Simulation: Simple Client Server Topology

- Client and Server connected with 1.5 Mbps line
- Packet discarder, between server to client link, impose packet loss
- File of size 3 MB is transferred from server to client using ftp application
- Different packet loss scenario is created for 0.5 second



Client server topology

# No packet loss

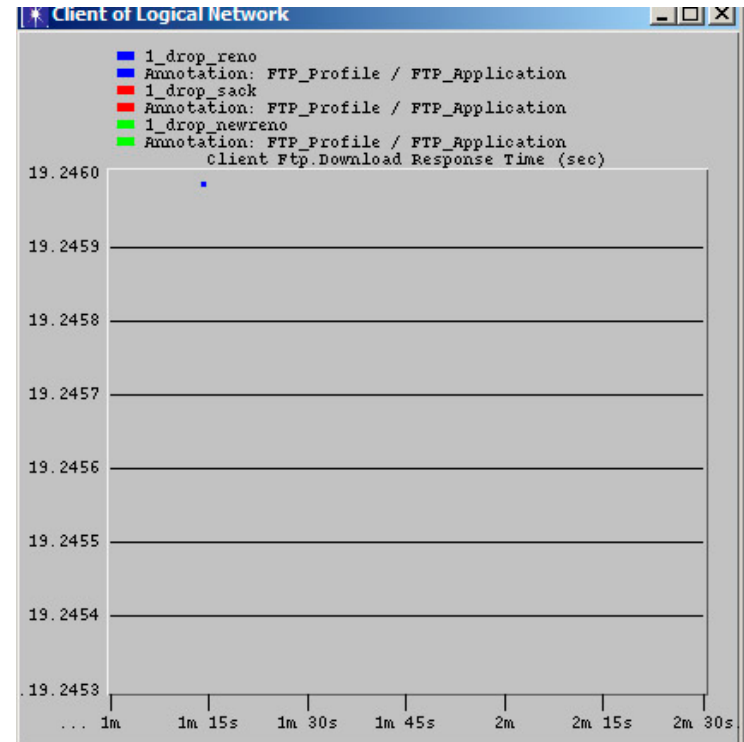
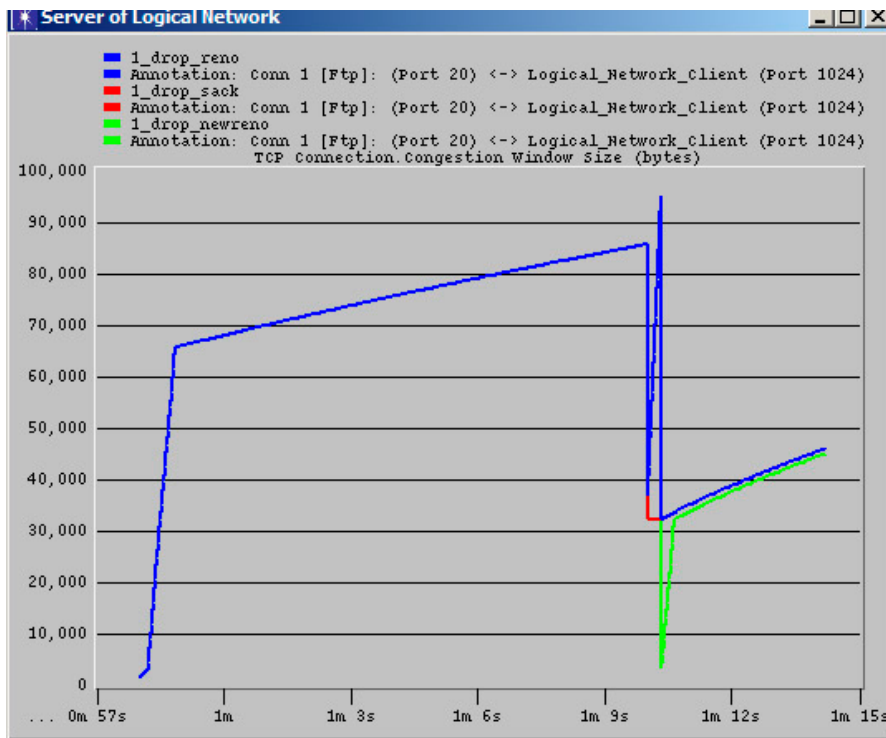


Congestion window (Reno, SACK, NewReno)

Download response time



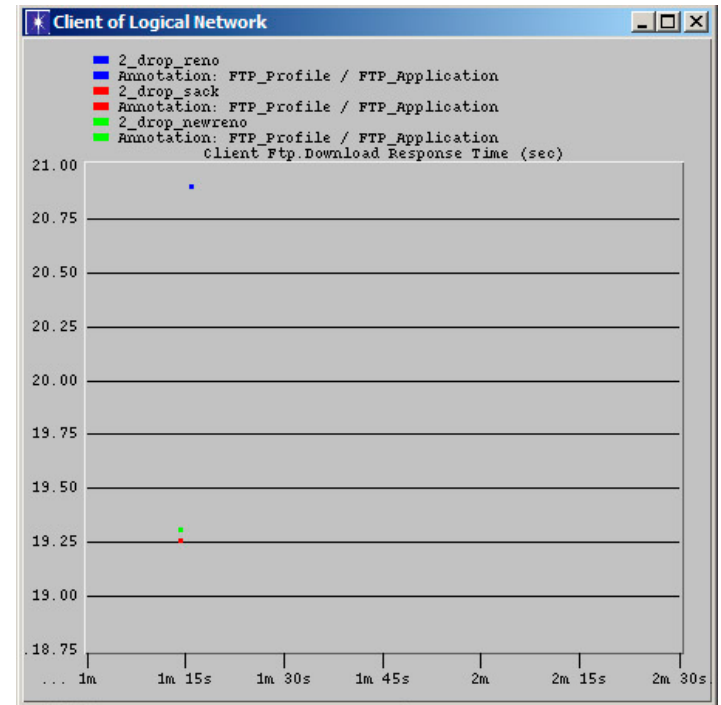
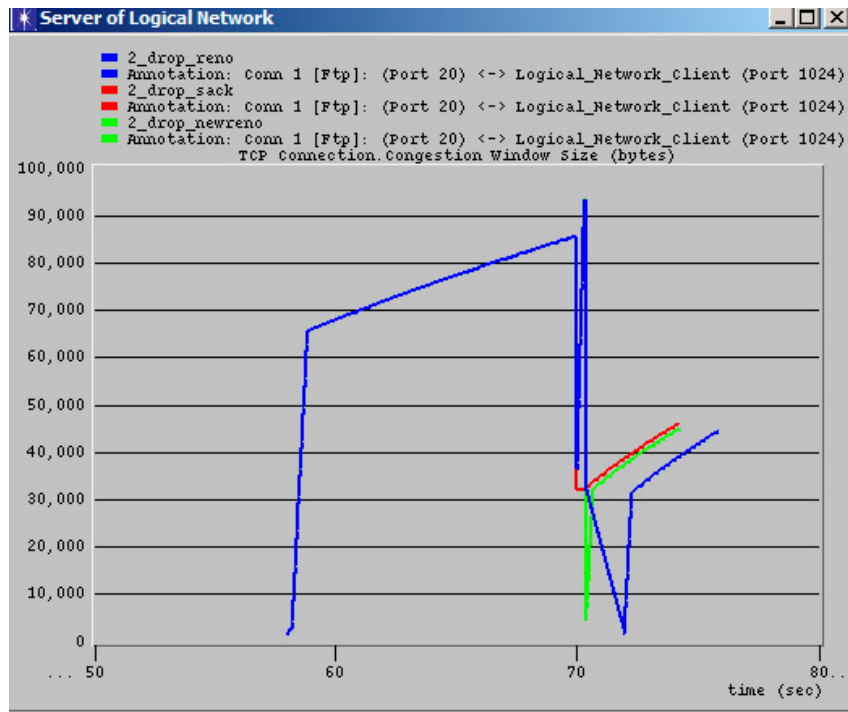
# One packet loss



Congestion window (Reno, SACK, NewReno)

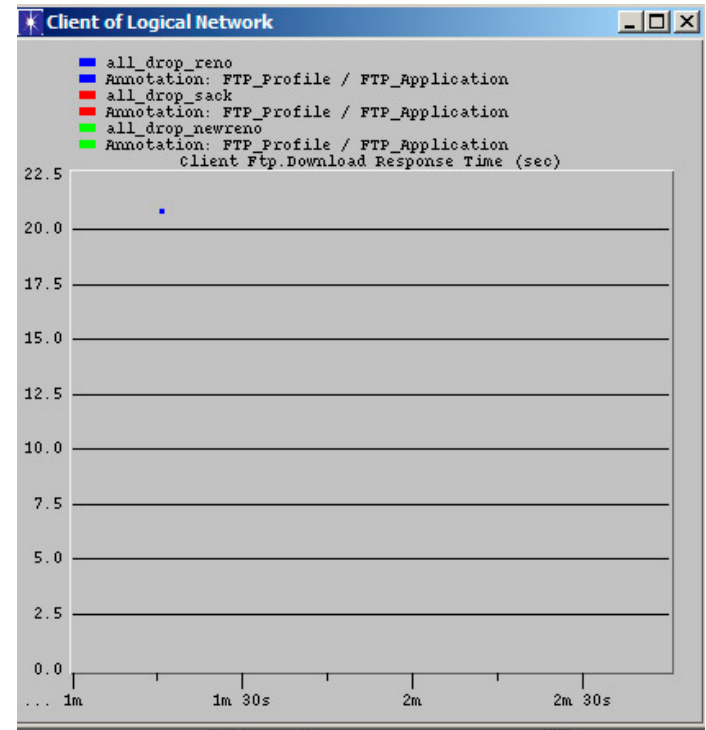
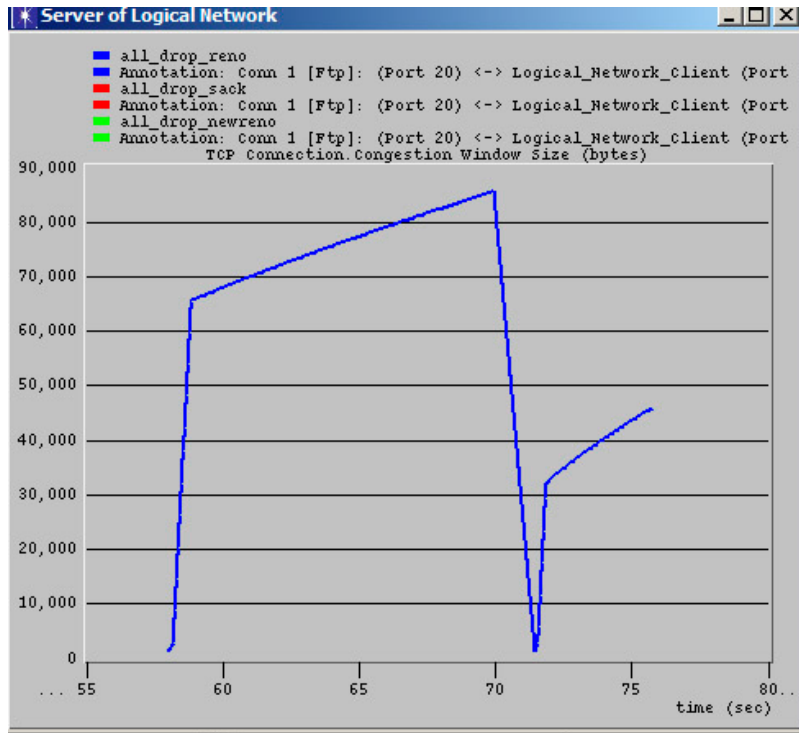
Download response time

# Two packets loss



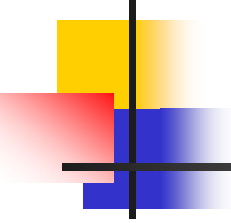
Congestion window (Reno, SACK, NewReno) Download response time

# All packets loss



Congestion window (Reno, SACK, NewReno)

Download response time

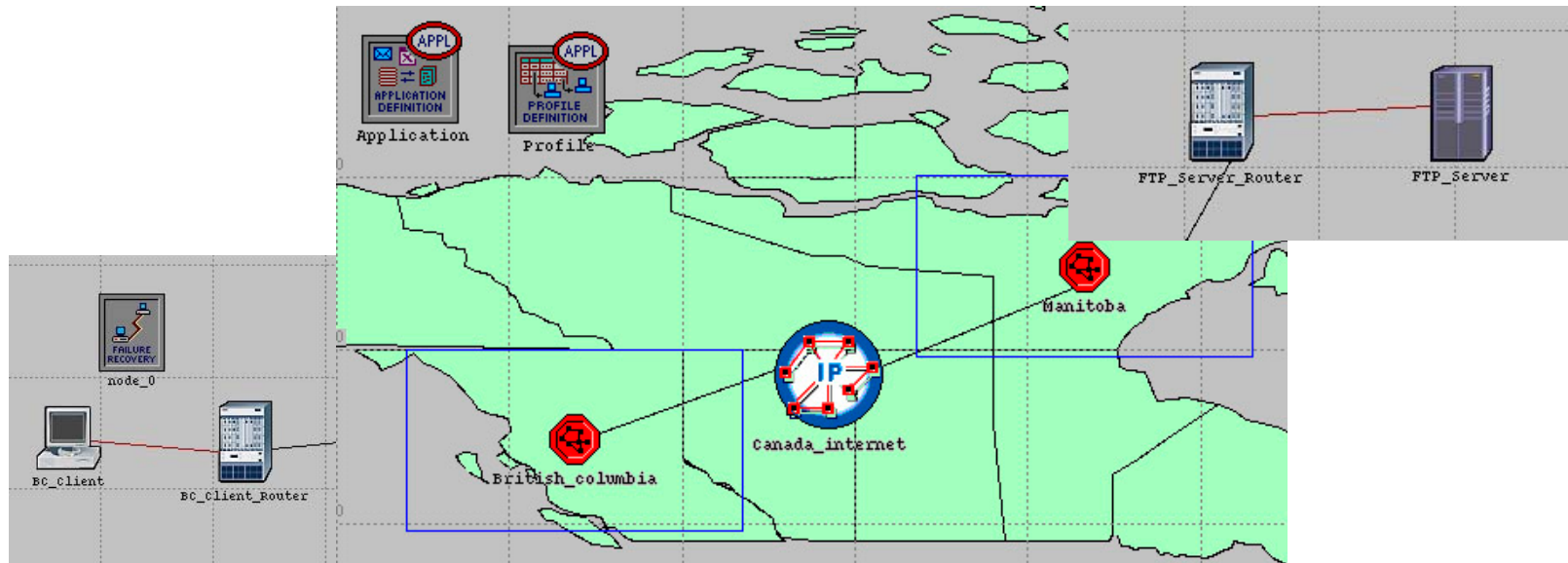


# Simulation: Client and Server with disconnection node

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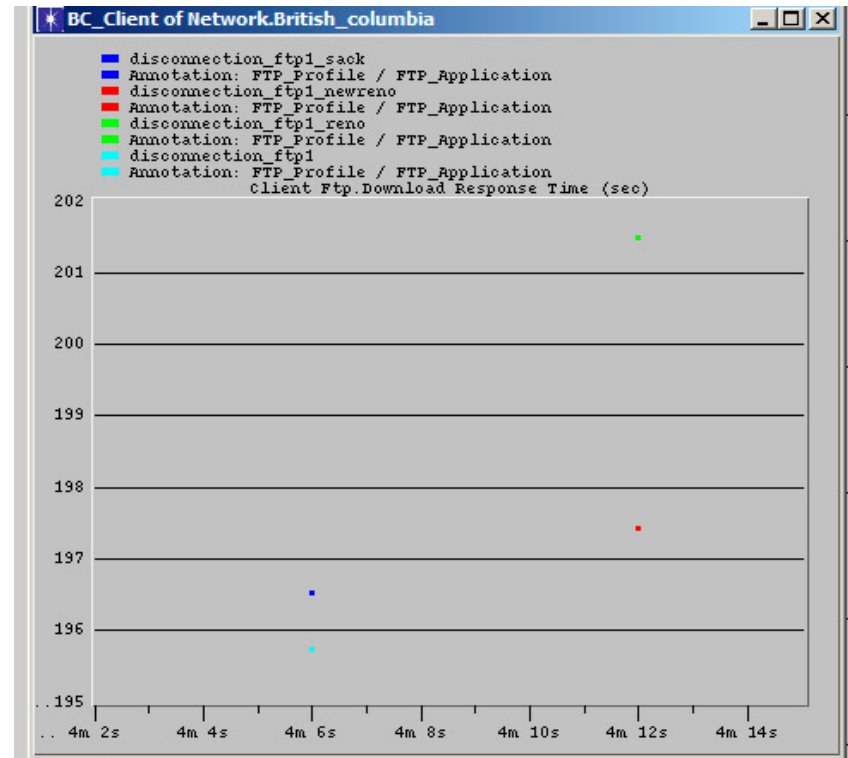
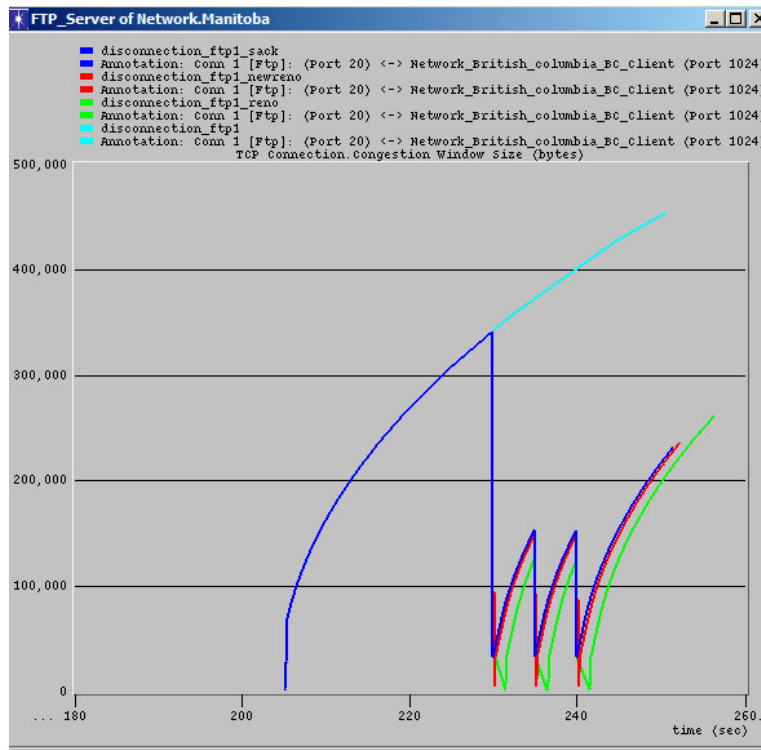
- consist of two subnets: Client and Server in each
- client and server connected to routers: 100 Mbps link
- routers are connected to the Internet cloud: 45 Mbps link
- 150 MB is transferred using ftp application
- link between client and router is disconnected via failure recovery node
- disconnection time intervals:
  - 0.05s, 0.1s, and 0.2s
  - 10s

# Simulation: Client and Server with disconnection node



Network topology to simulate disconnected network

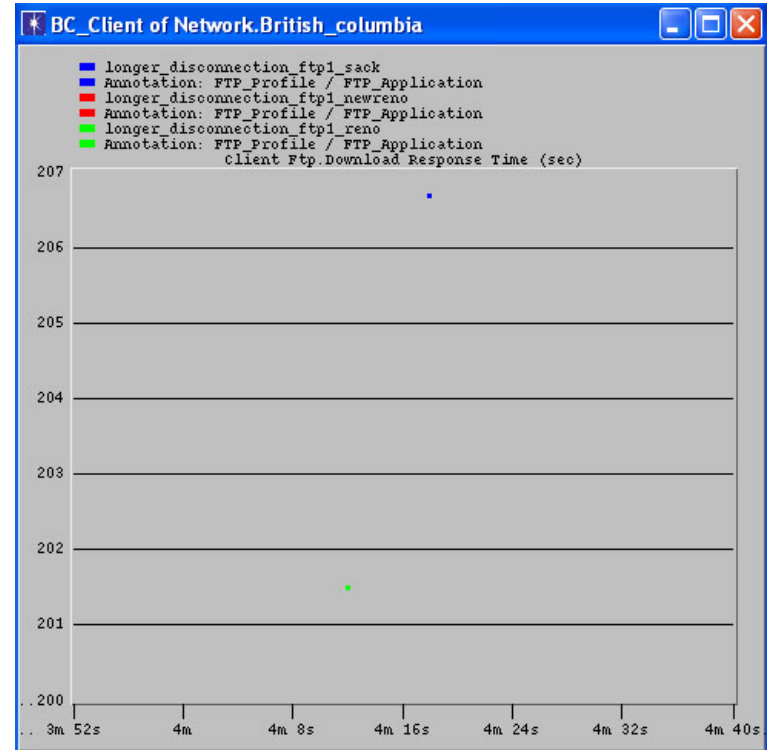
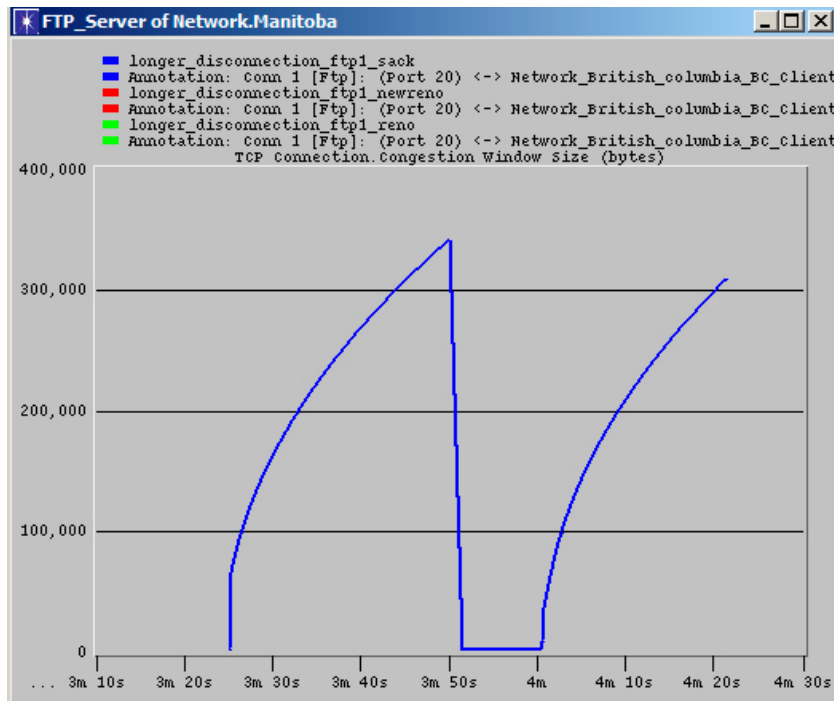
# Disconnection for 0.05s, 0.1s, 0.2s



Congestion window (SACK, Reno, NewReno)

Download response time

# Disconnection for 10s



Congestion window (Reno, SACK, NewReno)

Download response time



# Simulation: Client and Server with congested network

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- consists of multiple subnets: multiple clients and servers
- client and server connected to each subnet router by 100 Mbps link
- 45 Mbps link connects routers to backbone Internet
- database and ftp application are used in server
- profile using server application are run on client
- congested network is simulated



# Simulation: Client and Server with congested network

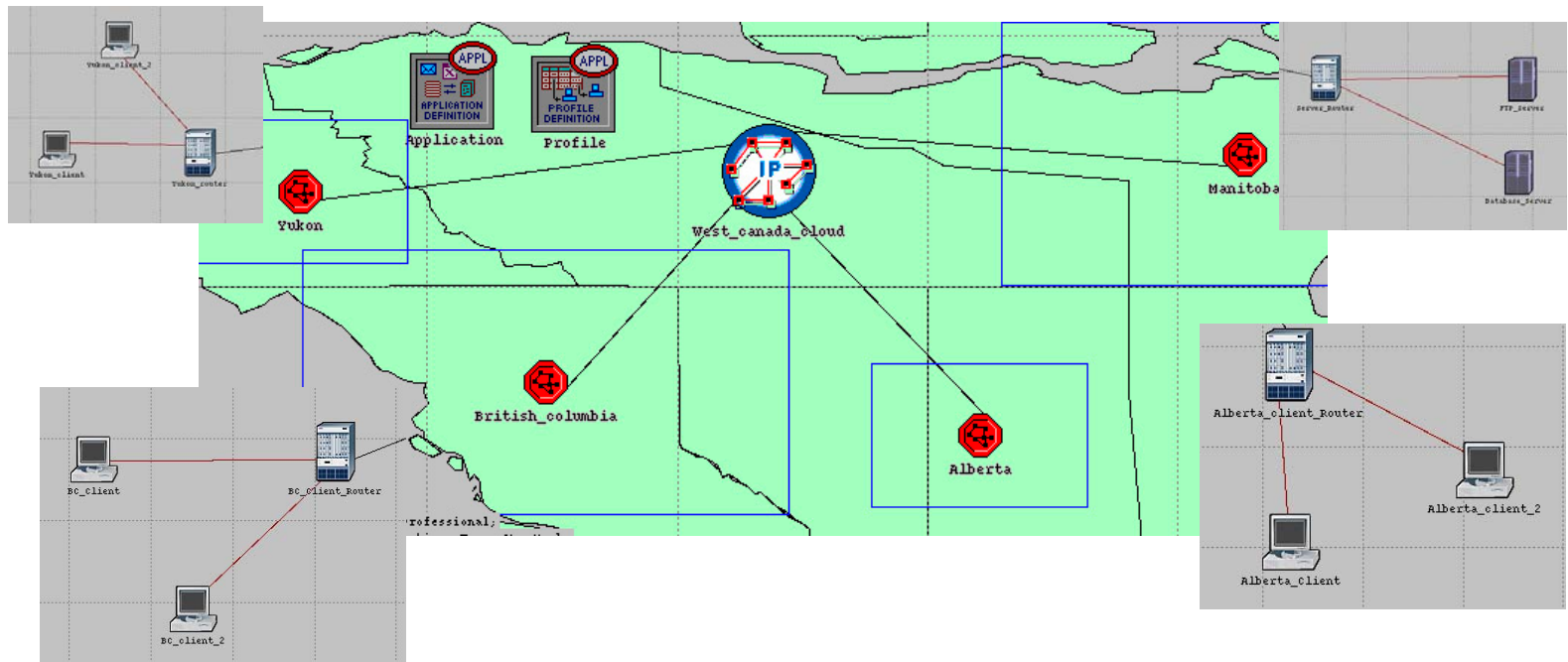
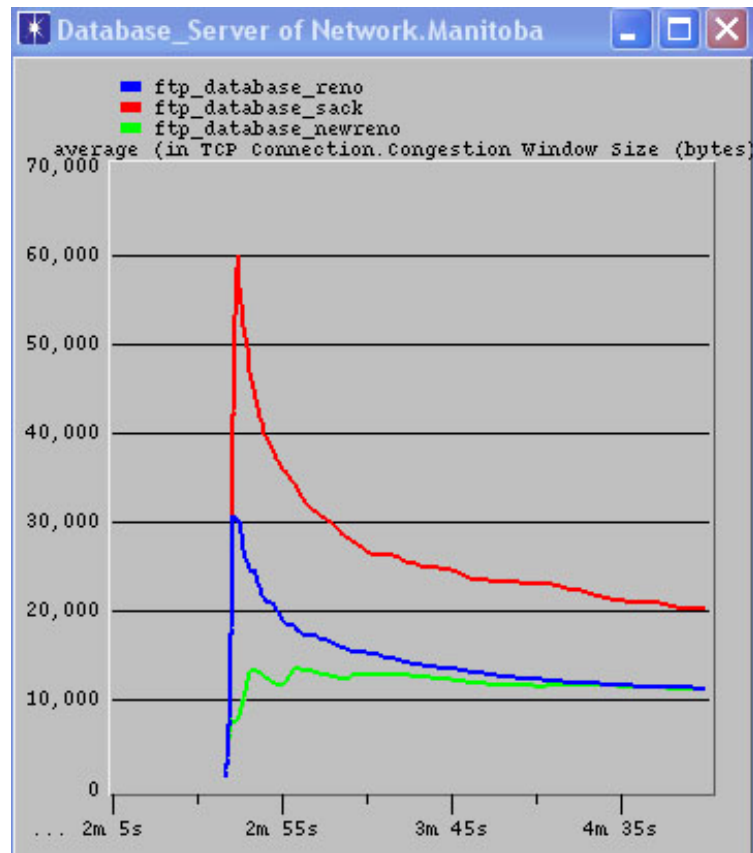


Fig: Network topology used to simulate congested network

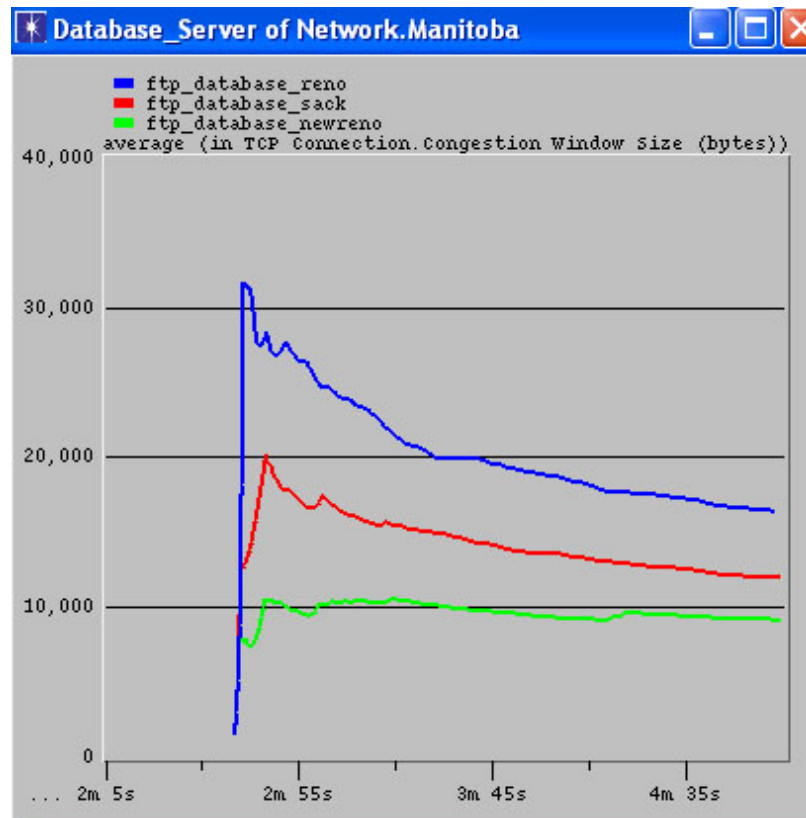
# Congestion window at port 1025



Average congestion window at port 1025

Reno, Sack, and New-Reno

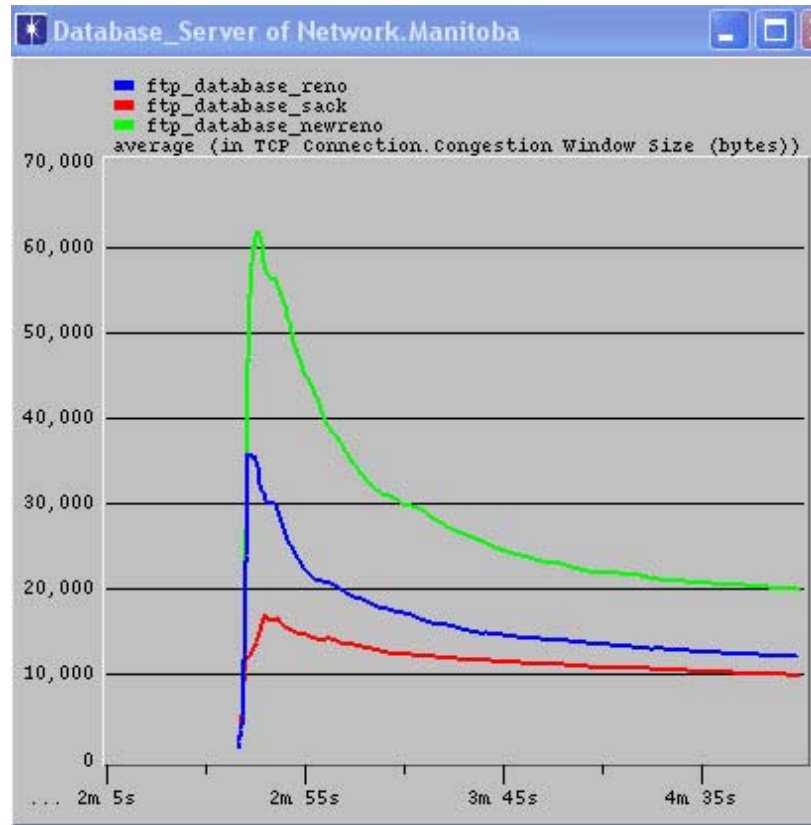
# Congestion window at port 1026



Average congestion window at port 1026

Reno, Sack, and New-Reno

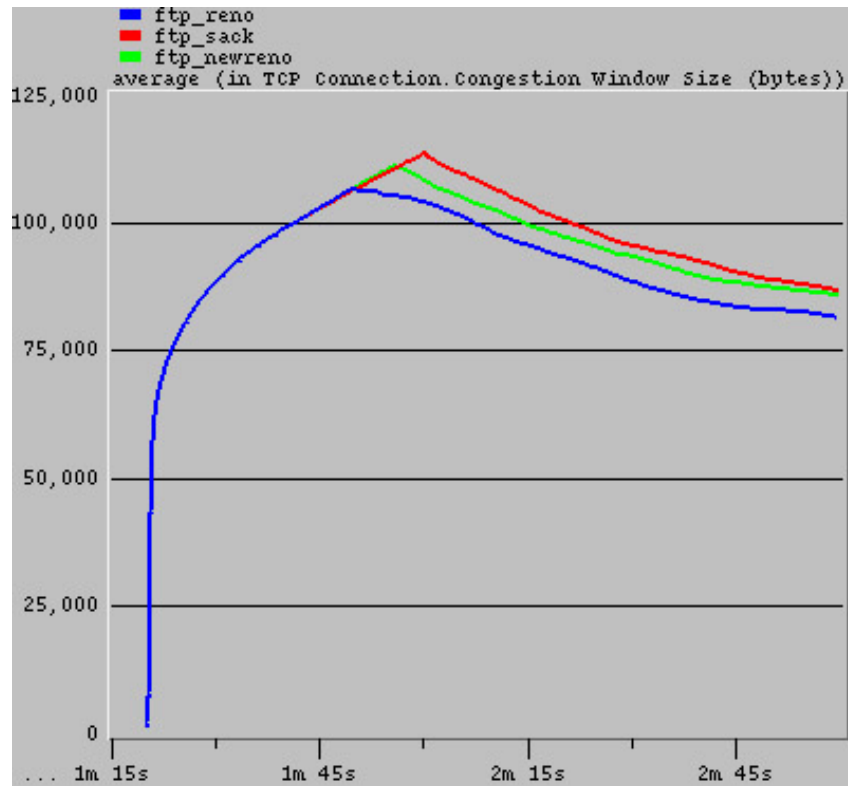
# Congestion window at port 1027



Average congestion window at port 1027

Reno, Sack, and New-Reno

# Slightly congested network



Average congestion window for low congested network

Reno, Sack, and New-Reno



# Conclusion

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- Simple client server model:
  - Reno, SACK, and NewReno recovers congestion window similarly in case of all and no packets loss
    - SACK performs best for multiple packets loss
    - Reno performs worst among three algorithms
- Client and server with disconnected node:
  - All algorithms are incapable to distinguish link disconnection



# Conclusion

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- window recovery process varies with disconnection interval
- SACK performs better for short disconnection
- **Multiple clients and servers model:**
  - conflicting behavior for heavily congestion network
  - SACK performs better in slightly congested network



# Future Work

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- model wireless node with disconnection behavior for detailed analysis
- analyze variation in drop time for more further performance analysis
- investigate to identify the cause of conflicting behavior of Reno, NewReno, and SACK algorithms for heavily congested network
- implement new algorithms and compare their performance with these basic algorithms





# References

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- [1] S. Floyd and K. Fall, "Simulation Based Comparisons of Tahoe, Reno and Sack TCP", ACM Computer Communication Review, 1996, Vol.26, No.3: 5–21 [1]
- [2] S.Floyd and T.Henderson "The New-Reno Modification to TCP's Fast Recovery Algorithm" RFC 2582, Apr 1999. [2]
- [3] R. Paul and Lj. Trajkovic, "Selective-TCP for wired/wireless networks," Proc. SPECTS 2006, Calgary, AL, Canada, Aug. 2006, pp. 339-346. [3]
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- [5] Z. Chen and M.M. Ali, "The performance of TCP congestion control algorithm over high-speed transmission links," IEEE Canadian Conf., Department of Electrical and Computer Engineering, Concordia University, Montreal, Canada, May 2004, pp.1371 - 1374 Vol.3. [5]



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- [6] G. Holland and N. Vaidya, "Analysis of TCP performance over mobile ad hoc networks," Proc. ACM/IEEE Int. Conf. on Mobile Computing, Networking,, Seattle, Washington, United States, 1999, pp. 219-230.[6]
- [7] F. Anjum and L. Tassiulas, "Comparative study of various TCP versions over a wireless link with correlated losses", *IEEE/ACM Transactions on Networking (TON)*, NJ, USA, June 2003, Vol. 11, Issue 3, pp. 370 – 383. [7]
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- [9] RFC [online]. Available: <http://www.ietf.org/rfc.html>.