CSE 232: Programming Assignment 4 Network simulation and TCP congestion control analysis using ns3

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Question 1. Run the simulation with the default parameters (provided in the table) and answer the following questions.

a.) What is the maximum expected value (theoretical) of throughput (in Mbps)? Why?

The maximum throughput of the network is equivalent to the bottleneck bandwidth of the network. The bandwidth between the nodes NO and N1 is 10 Mbps and the bandwidth between the nodes N1 and N2 is 7 Mbps. The bottleneck bandwidth will be the minimum of both these bandwidths. Hence, the maximum value of expected throughput is **7 Mbps**.

b.) How much is Bandwidth-Delay-Product (BDP)? Express your answer in terms of the number of packets.

We know that

$$BDP = Bandwidth \times End-to-end delay$$

Here, the bandwidth is equivalent to the bottleneck bandwidth (7 Mbps in this case) and end-to-end delay is the total round trip time for the packet between the nodes NO and N2 through N1 and the transmission delay between them.

End-to-end delay = RTT between
$$N_0$$
 and $N_2 + 2 \times d_{trans}$
= $2 \times (\text{delay between } N_0 \text{ and } N_2) + \frac{2 \times 1460 \times 8}{7 \times 10^6}$
= $2 \times (\text{delay between } N_0 \text{ and } N_1 + \text{delay between } N_1 \text{ and } N_2)$
= $2 \times (100 + 10) = 2 \times 110$
= $223.33 \,\text{ms}$

Hence,

BDP = 7Mbps
$$\times$$
 223.33ms
= 1563.31 \times 10⁶ \times 10⁻³bytes
 \approx 1.56Mb

Application Payload size = 1460 bytes = 11680 bits.

$$\begin{aligned} \text{Number of packets} &= \frac{\text{BDP}}{\text{Payload size}} \\ &= \frac{1.56 \times 10^6}{11680} \approx 133 \, \text{packets} \end{aligned}$$

Hence, the BDP in terms of number of packets is 133 packets.

c.) What is the average computed throughput of the TCP transfer?

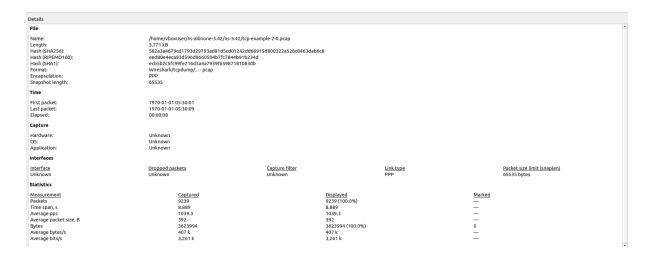


Figure 1: Received Packet

Bytes sent =
$$3623994$$

Time taken = 8.889 seconds
Throughput = $\frac{\text{Bytes sent}}{\text{Time taken}} = \frac{3623994 \times 8}{8.889 \times 10^6}$
 $\approx 3.262 \,\text{Mbps}$

Hence, the average computed throughput of the TCP transfer is **3.262 Mbps**.

d.) Is the achieved throughput approximately equal to the maximum expected value? If it is not, explain the reason for the difference.

No, the achieved throughput (3.262 Mbps) is not equal to the maximum expected value (7 Mbps). The achieved throughput can be less than the expected throughput due to the following possible reasons:

- Packet retransmissions: The sender might have to retransmit packets due to lost ACKs or packets.
- Network Congestion: The size of the receiver buffer might not be large enough, hence leading to the dropping of packets due to overflowing buffers caused by network congestion.
- End-to-end delay: The end-to-end delay between receiver and sender could be one of the reasons leading to inefficient utilization of the bandwidth available.

e.) Plot Congestion Window (CWND) with time.

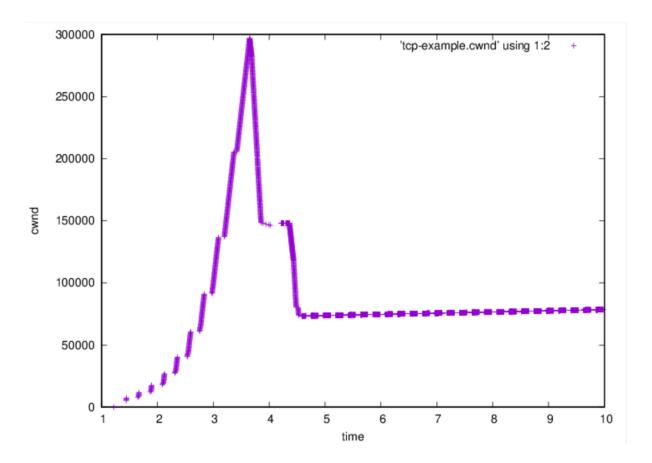


Figure 2: Congestion Window vs Time

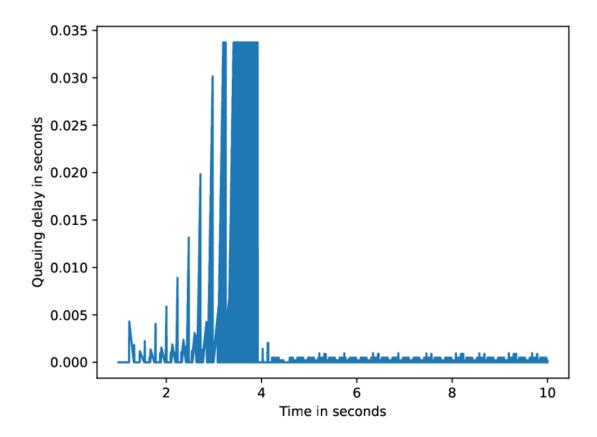


Figure 3: Queueing Delay vs Time

g.) Are the plots in 1(e) and 1(f) related?

Yes, the plots in 1(e) and 1(f) are related.

We notice that as the congestion window increases exponentially during the slow start phase, the queueing delay also increases correspondingly as more packets are being sent over the packet as the congestion window increases, leading to higher queueing delays.

As the congestion window drops around 4 seconds, likely due to a congestion event, the queueing delay also drops significantly, indicating that fewer packets are now being sent as congestion avoidance.

After the congestion is controlled, the network reaches a congestion avoidance phase where the congestion window is now stably increasing linearly with time. The queueing delay has also become stable, indicating reduced congestion.

Question 2. Change queue size to 1000.

a.) What is the average computed throughput of the TCP transfer?

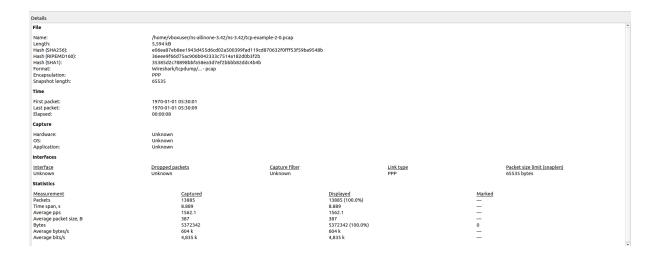


Figure 4: Received Packet

Bytes sent =
$$5372342$$

Time taken = 8.889 seconds
Throughput = $\frac{\text{Bytes sent}}{\text{Time taken}} = \frac{5372342 \times 8}{8.889 \times 10^6}$
 $\approx 4.835 \,\text{Mbps}$

Hence, the average computed throughput of the TCP transfer is ${\bf 4.835~Mbps}.$

b.) Plot CWND with time.

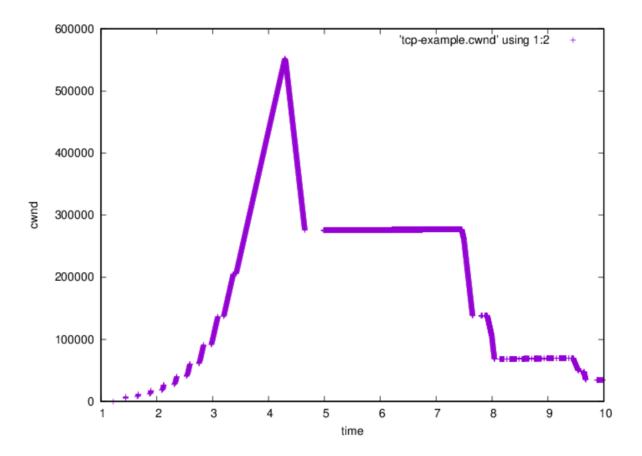


Figure 5: Congestion Window vs Time

c.) Plot queueing delay with time.

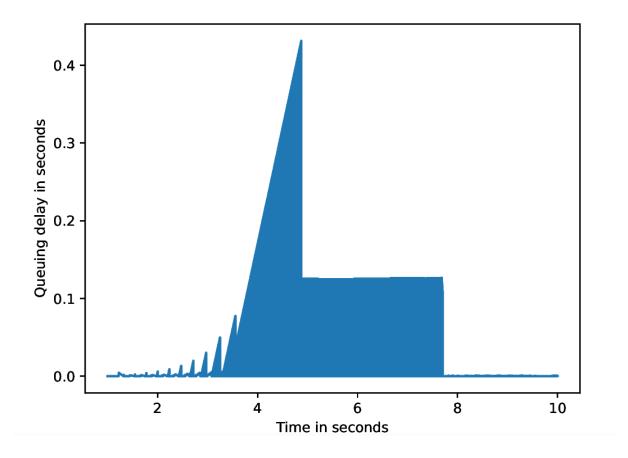


Figure 6: Queueing Delay vs Time

d.) Compare CWND plots of Q.1. and Q.2.; what insights did you gain?

From the CWND plots of Q1 and Q2, we notice that the congestion window of plot 2 increases for a longer time and reaches a higher peak than plot 1. This is due to increased queue size, which allows for a larger buffer, hence allowing the sender to send more packets over the network and the overflow does not happen quickly. Due to the higher congestion window, we notice more drops in size as compared to plot 1 during the congestion avoidance phase.

Question 3. Change N1-N2 bandwidth to 10 Mbps and N1-N2 delay to 100ms.

a.) What is the average computed throughput of the TCP transfer?

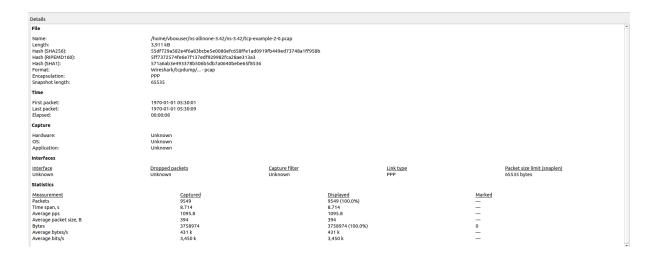


Figure 7: Received Packet

Bytes sent =
$$3758974$$

Time taken = 8.714 seconds
Throughput = $\frac{\text{Bytes sent}}{\text{Time taken}} = \frac{3758974 \times 8}{8.714 \times 10^6}$
 $\approx 3.451 \text{ Mbps}$

Hence, the average computed throughput of the TCP transfer is $\bf 3.451~Mbps.$

b.) Plot CWND with time.

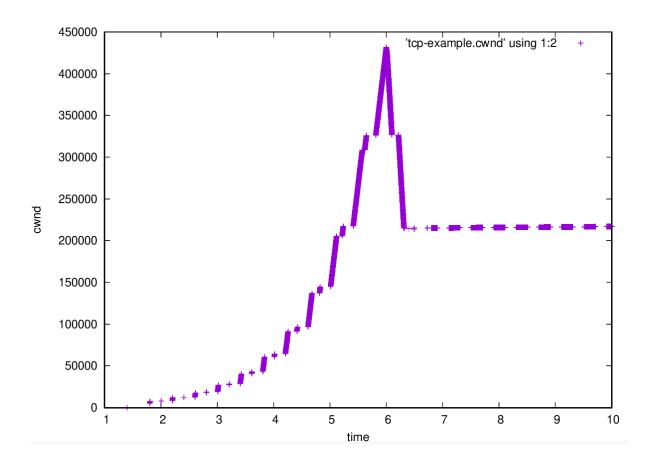


Figure 8: Congestion Delay vs Time

c.) Plot queueing delay with time.

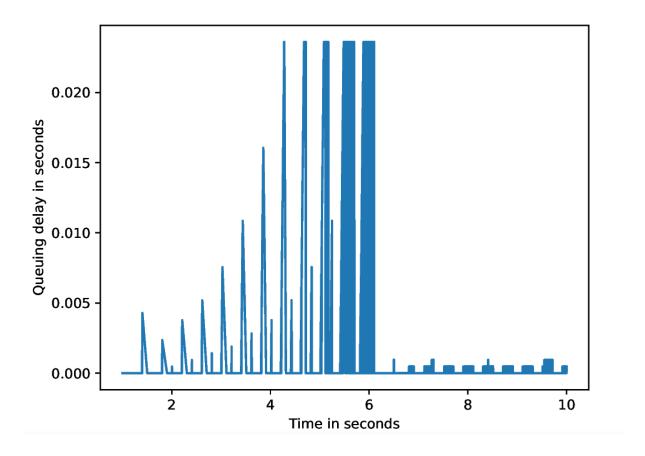


Figure 9: Queueing Delay vs Time

d.) Compare queuing delay plots of Q.1. and Q.3.; what insights did you gain?

By comparing queueing delay plots of Q1 and Q3, we gain the following insights:

- Since there is a mismatch in the bandwidth between the nodes in the first case, it leads to a buildup of the queue as the transmission rate of 7Mbps is slower than the arrival rate. Hence, the peak of queueing delay is higher in case 1 compared to case 2 and is reached comparatively faster.
- There is a delay mismatch between both the links in the first case so it further leads to queue buildup too. When the delay is similar in second case, it smoothens the queueing delay curve due to delayed ACKs and transmissions.

We can conclude that the enqueue time is higher in the first case as compared to dequeue time leading to higher delays.

References

- [1] GitHub
- [2] GeeksForGeeks: TCP Congestion Control
- [3] NS3: TcpSocket Class Reference