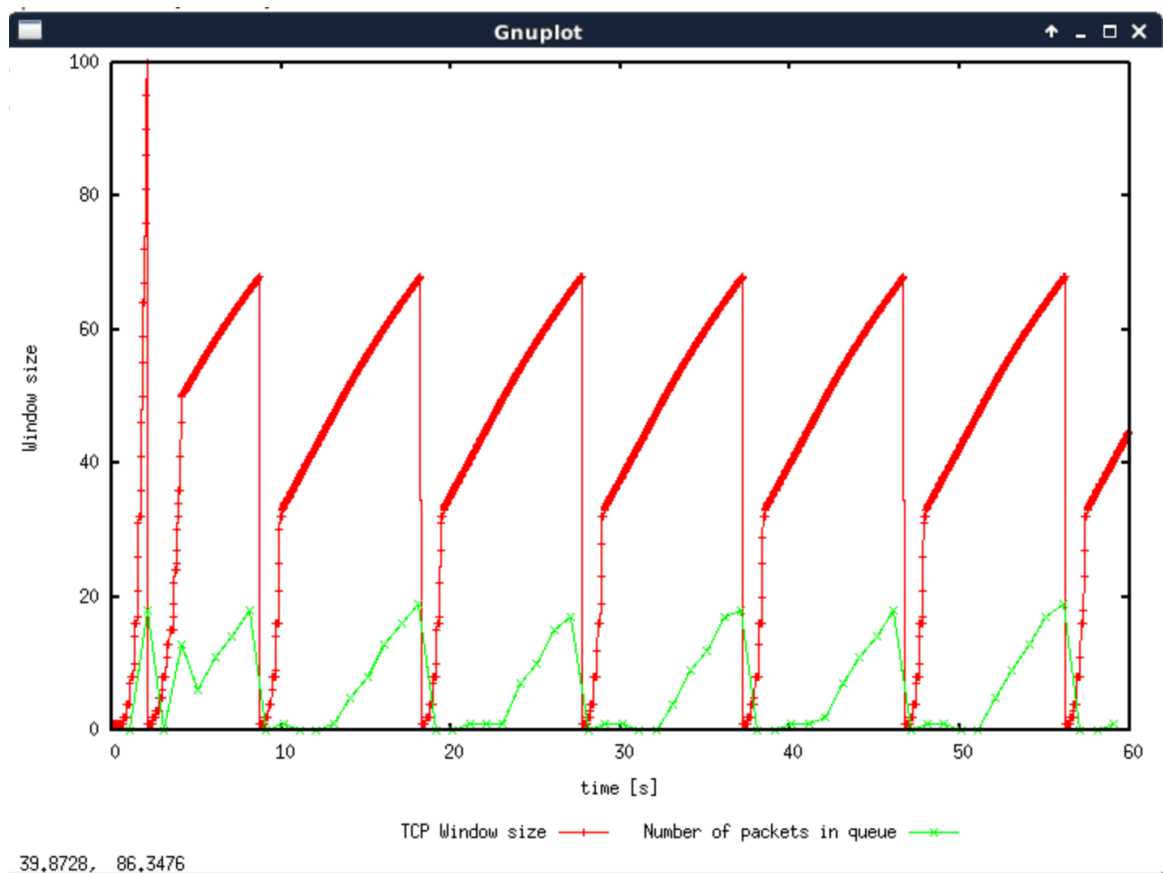


COMP9331 LAB 05

YUAN GAO Z5239220

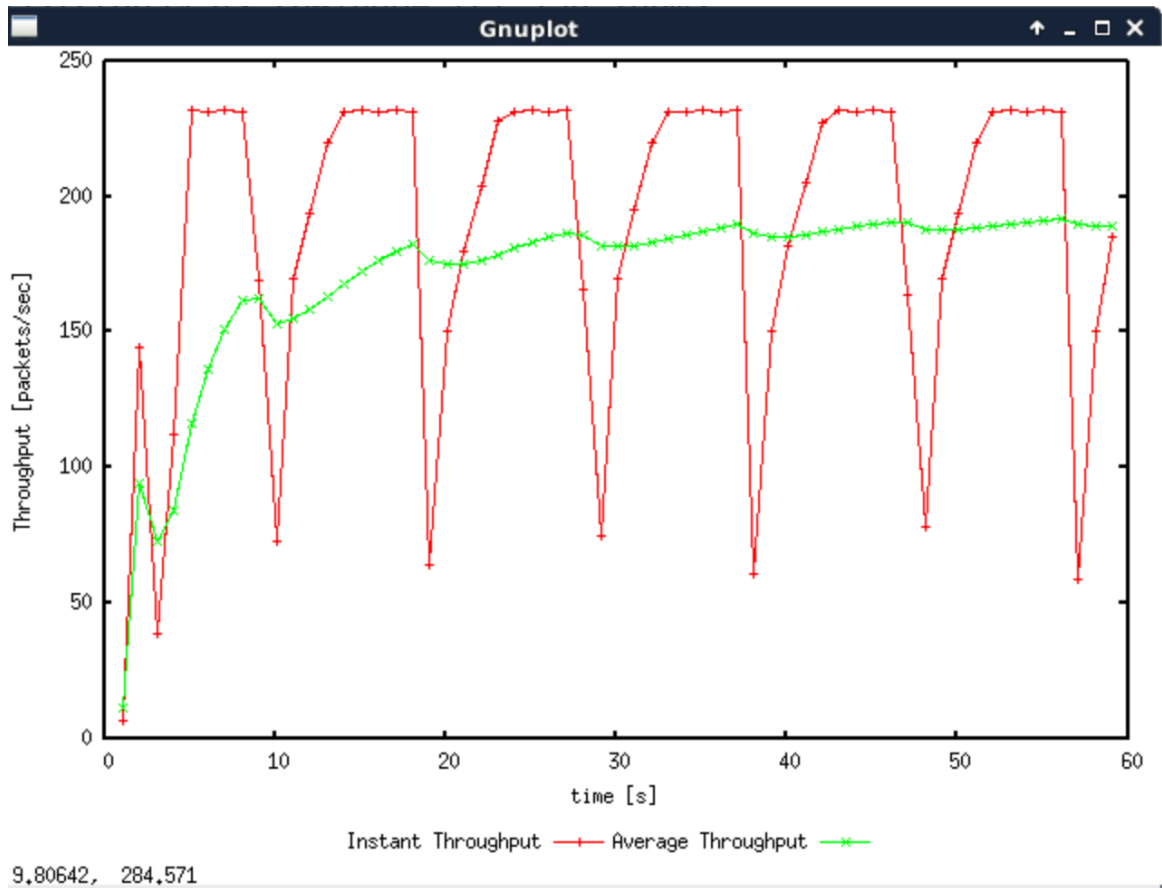
Exercise 1: Understanding TCP Congestion Control using ns-2

Question 1



- The maximum size of the congestion window that the TCP flow reaches in this case is 100
- Because the maximum size of queue is 20, any additional packets which more than 20 will be dropped
- When it reached to 100MSS, the queue was full and the packets dropped which results in a congestion event. The sender stopped increasing the congestion window size and reseted the window size as 1 MSS, the threshold as 50 packets. Then, The connection enters slow start. When the congestion window reached threshold congestion avoidance phase such as AIMD was used to avoid congestion. Afterwards, repeat the process.

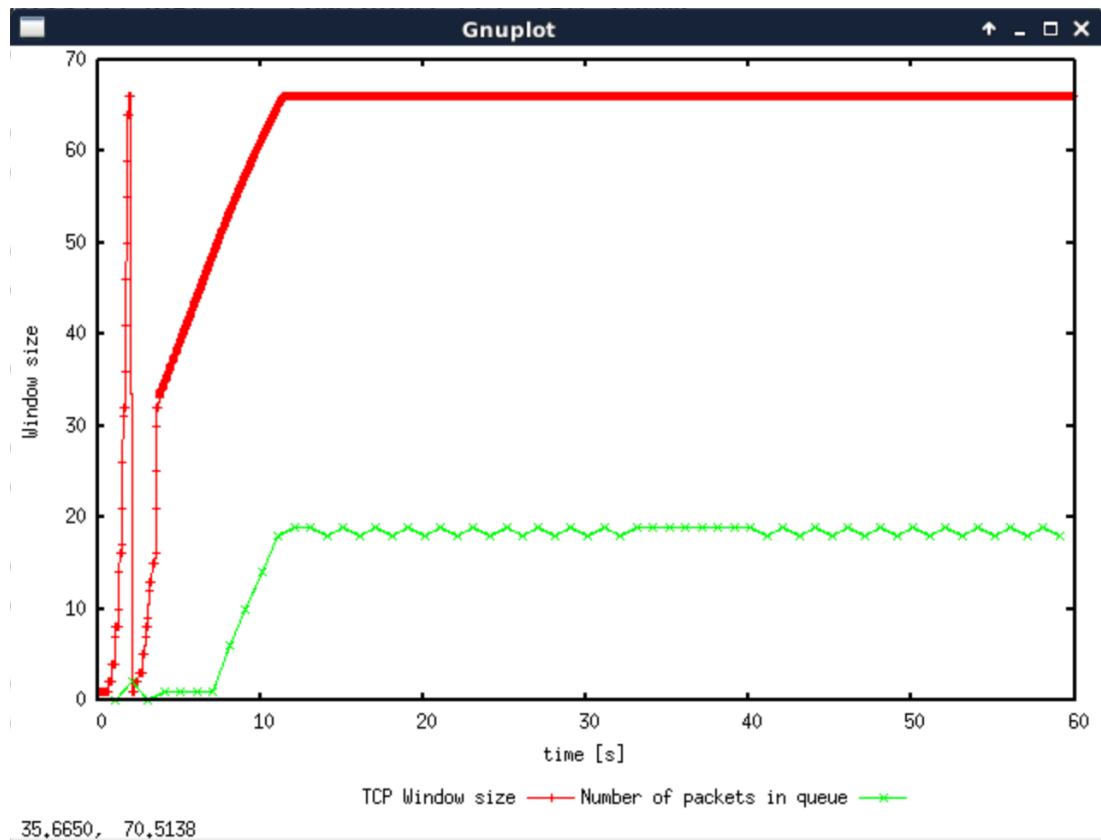
Question 2



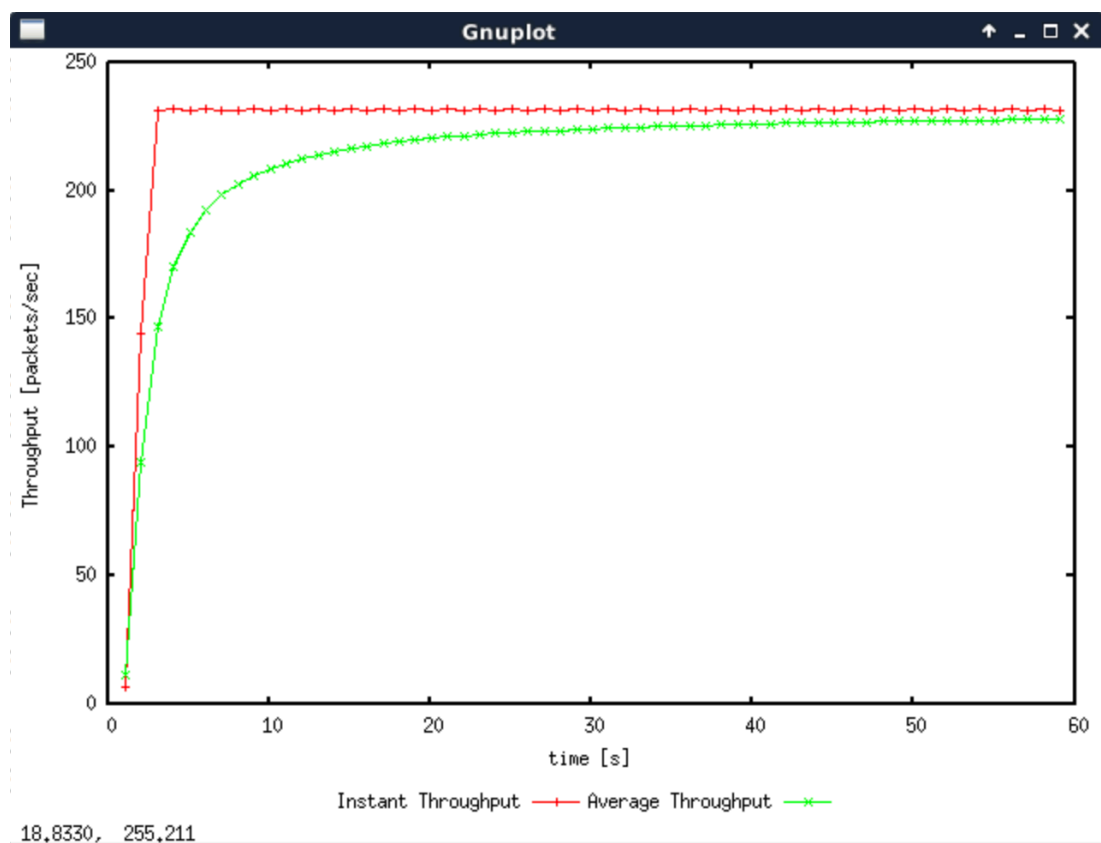
- The average throughput of TCP is around 190 pps.
- According to the graph, after 20s, the average throughput is relatively stable .
Therefore, The size of packets is $500 + 20 + 20 = 540 \text{ bytes}$
The throughput is $540 \times 190 \times 8 = 820.8 \text{ kbps}$

Question 3

- When the maximum congestion window size ≤ 66 , TCP stops oscillating to reach a stable behaviour. The window size stops after returning to the slow start for the first time. When we reduce the congestion window to half, this is enough to stabilize the number of packets in the send queue. The queue has never full, so these no packets are dropped.



- When the initial max congestion window is set to 50, TCP stabilized immediately.

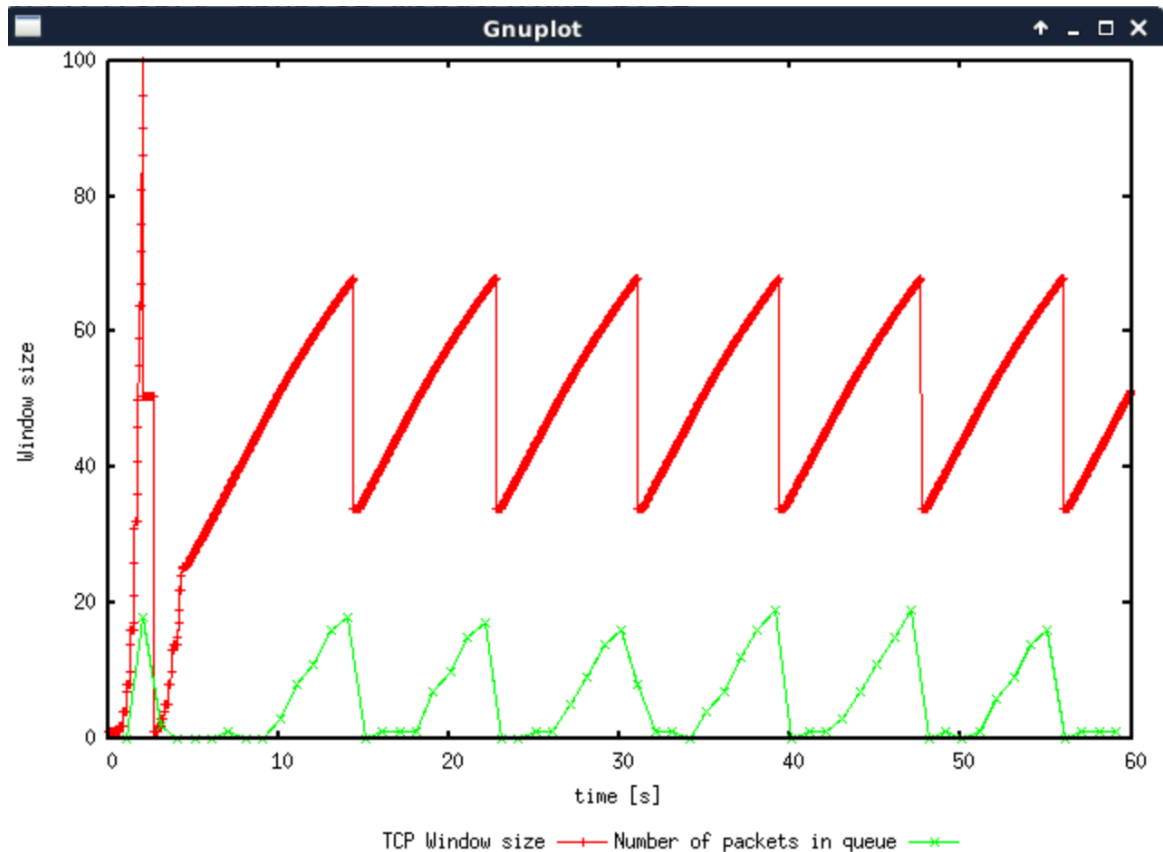


- The average packet throughput is around 225 pps.

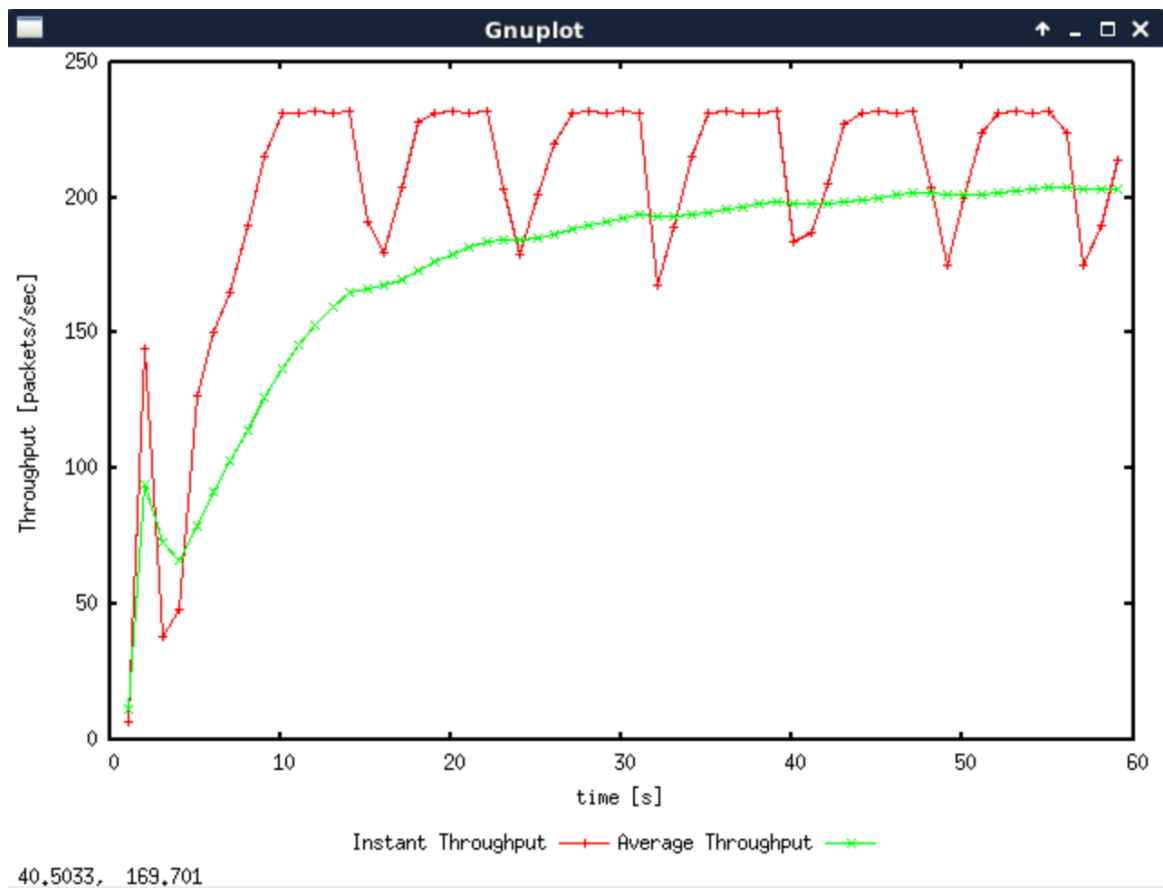
The throughput is $255 \times 500 \times 8 = 900 \text{ Kbps}$

It is almost equal to the link capacity.

Question 4

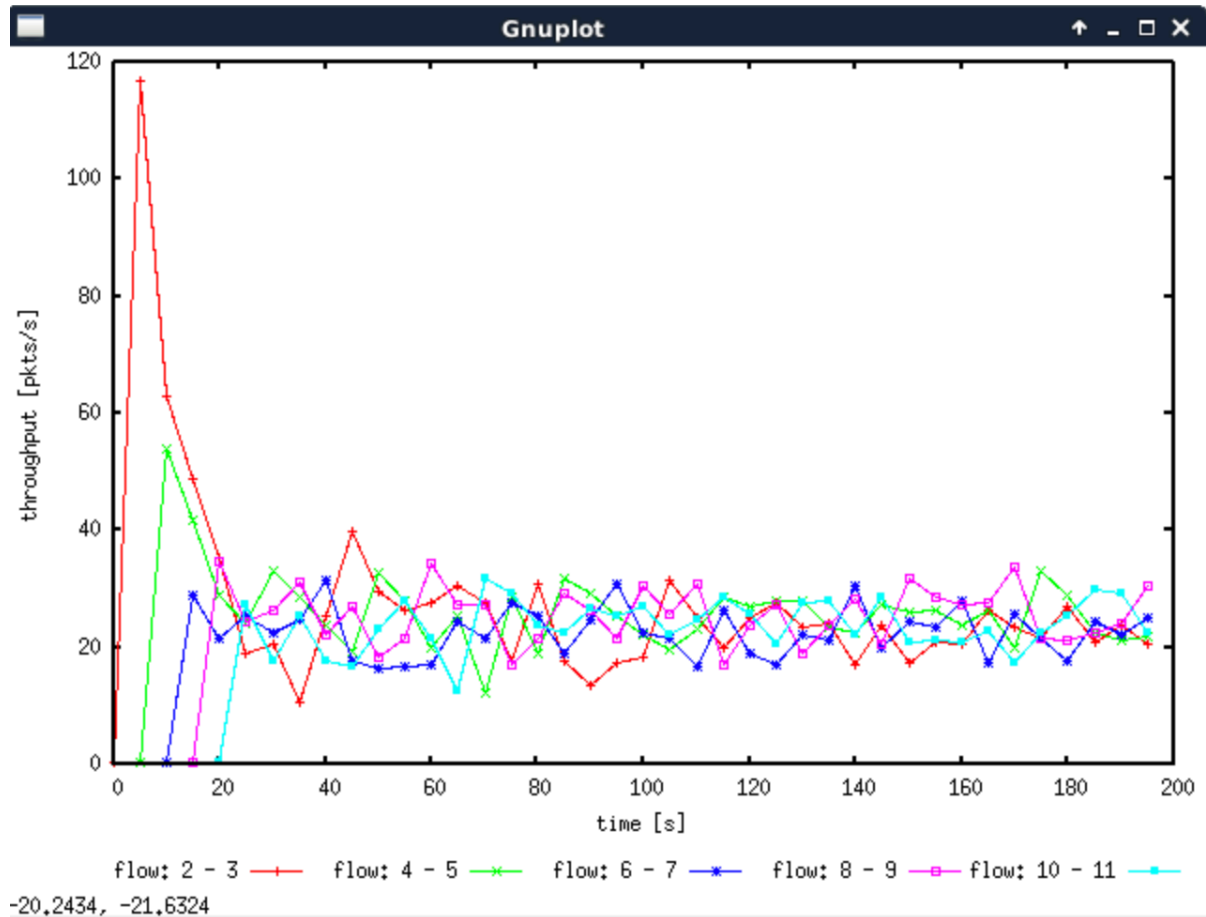


- The sender halves its current congestion window and increases it linearly, until losses. Afterwards, repeat the process. This because in TCP **Reno** method, most of the losses are detected due to triple duplicate ACKs. It is different to TCP **Tahoe** method. After each congestion event of TCP Tahoe, the window is reduced to 1.



- In TCP **Tahoe**, there are around 190 pps. The throughput is $540 \times 190 \times 8 = 820.8 \text{ kbps}$
- In TCP **Reno**, there are around 200 pps. The throughput is $540 \times 200 \times 8 = 863 \text{ kbps}$
- The throughput of **Tahoe** is slightly higher.

Exercise 2: Flow Fairness with TCP



Question 1

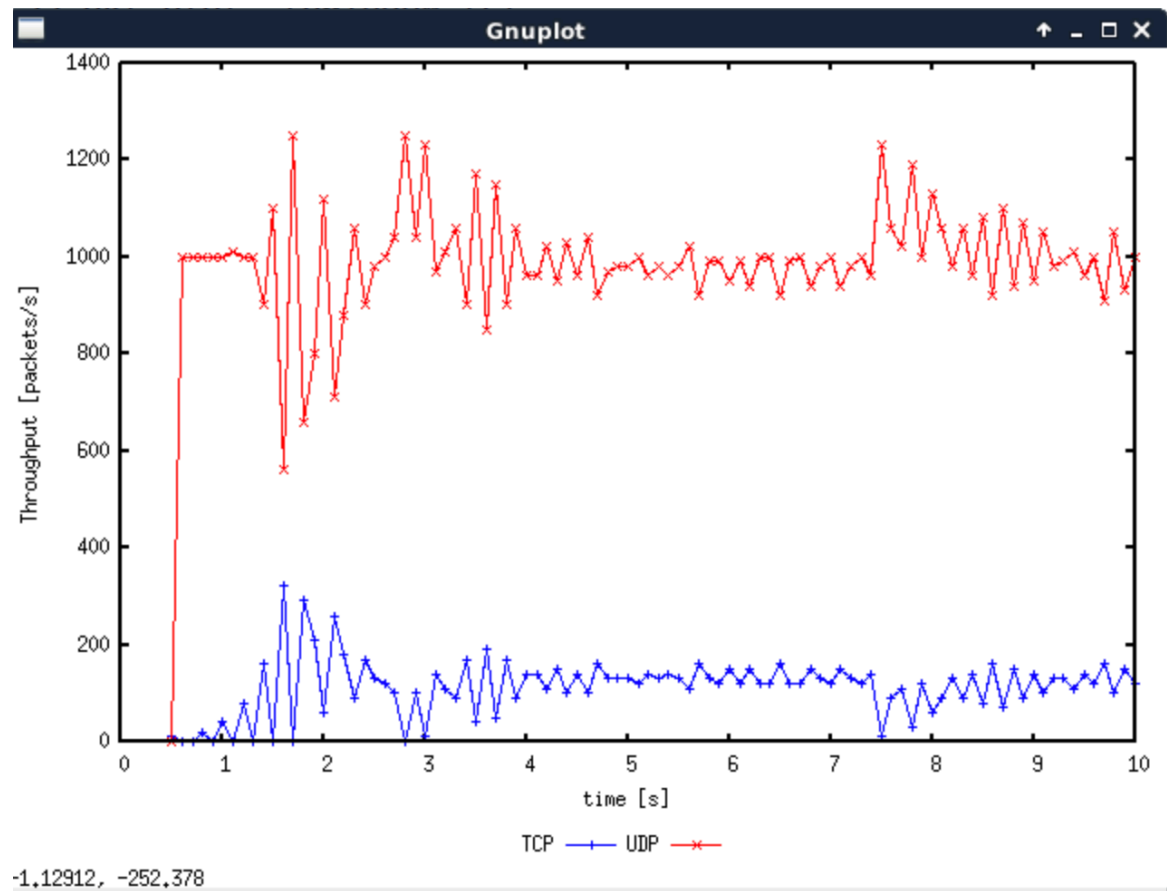
- Yes, each flow gets an equal share of the capacity of the common link. According to the graph, the five connections have commenced. These have similar throughput. Because when using the AIMD congestion control algorithm, window size can achieve long-term fairness.

Question 2

According to the graph, when a new flow is created, all go throughout pre-existing flows is reduced. This is because that during the slow start, when a new flow adds in, the new flow will increase quickly and result in congestion. Then, every TCP connection will adapt the size of its congestion window to avoid network congestion. This behaviour is fair, because once new traffic is joined in, all existing traffic is reduced accordingly.

Exercise 3: TCP competing with UDP

Question 1



- According to the picture, when the capacity of the link is 5 Mbps, the throughput of UDP is larger than TCP.

Question 2

This is because UDP does not have congestion control. Hence UDP will transmit packets keeping a constant rate regardless of the situation of packets dropped. However, TCP has congestion control, packets dropped affect rate. Therefore, TCP flows will be forced to use a lower throughput due to more aggressive UDP flows.

Question 3

- Advantages:
UDP does not have restrictions. It can keep transmitting at a higher speed. The delay would be low.
- Disadvantages:
UDP is unreliable. Therefore, if using UDP to transfer files, it needs a reliable application.
UDP will increase the burden of the network.
- If everybody started using UDP instead of TCP, the network may encounter congestion collapse.

