

Assignment 5

Wireshark assignment for Transport & Link layer

Harsh Agarwal
CS15BTECH11019

Transport Layer for Intranet

Analysis of downloading ubuntu-16.04.3-desktop-amd64.iso from intranet

<http://intranet.iith.ac.in/files/os/ubuntu-15.04-desktop-amd64.iso>

Server IP - 192.168.35.5:3128

My IP - 192.168.116.237:46534

Capture for 58 seconds

Command to filter this transaction :: **tcp.stream eq 0**

1)

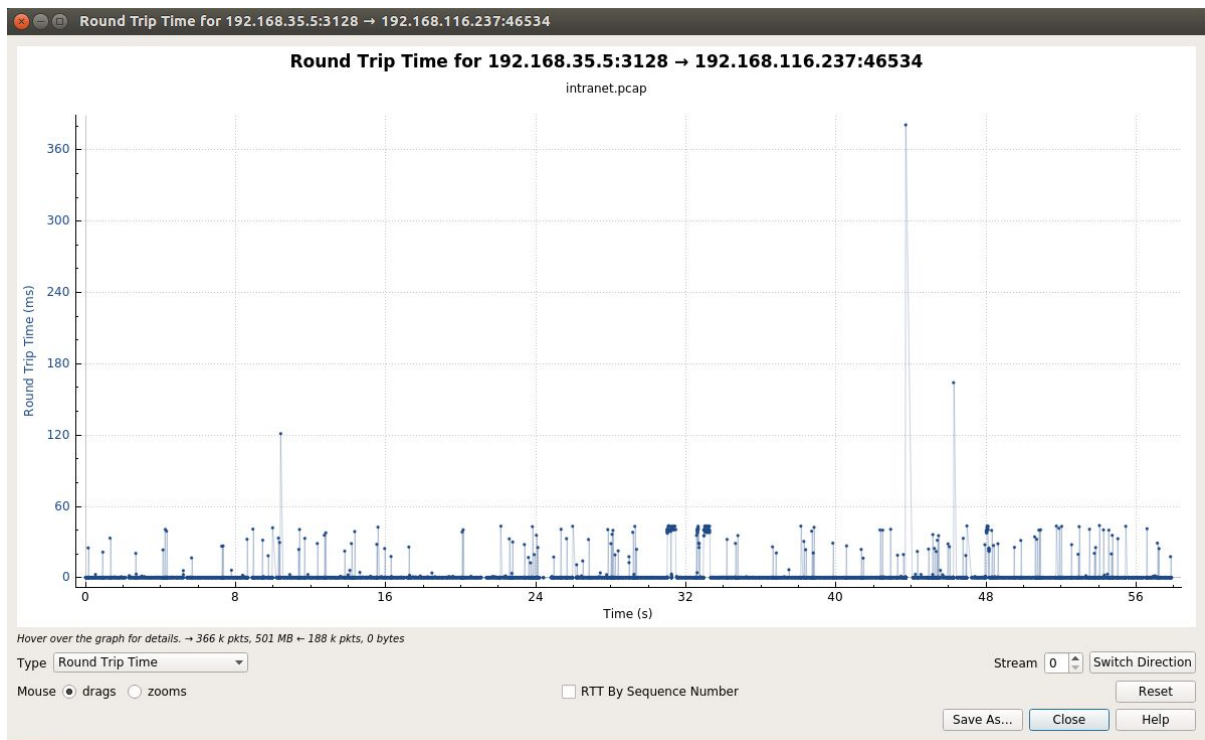
a) I have plotted RTT(in ms) vs Time(s) & RTT(in ms) vs Sequence Number.

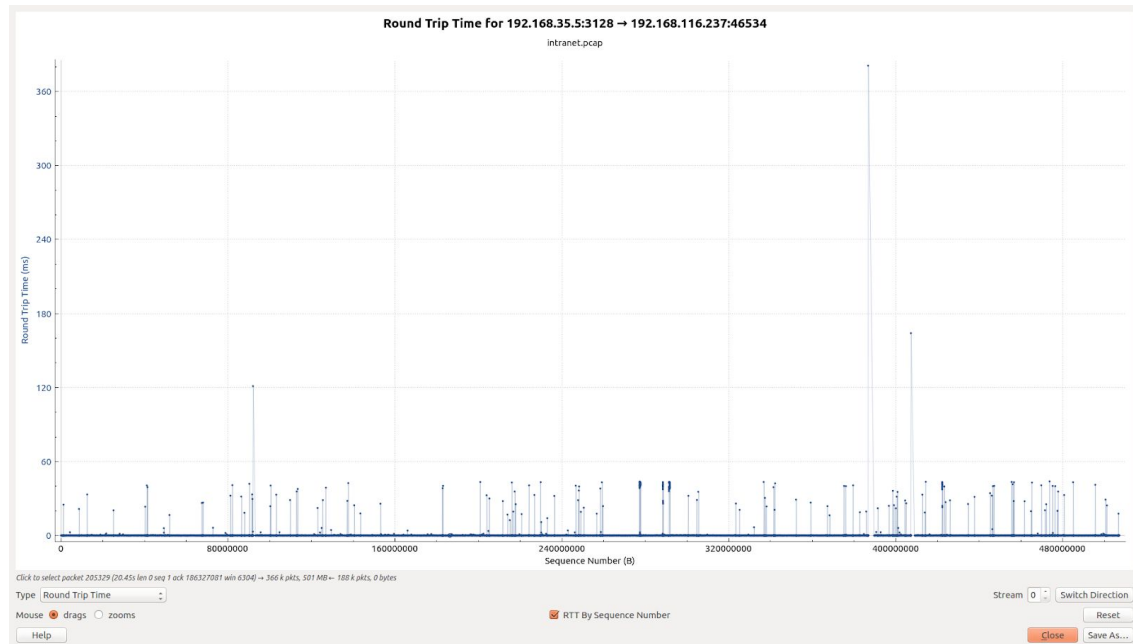
Most RTT values are close to 0, showing really less RTT.

There are a few fluctuations in the values with the maximum RTT being 380 ms.

Most of these fluctuation value are close to 50ms. There can be many reasons for these fluctuations.

Internet RTT was close to 10 ms, whereas intranet is close to 0ms. Since intranet is on a proxy server, this is much faster than internet(more proximity to the user).





b)

The IO Graph below is for `tcp.analysis.bytes_in_flight` vs time
`bytes_in_flight` tells how many bytes are now in flight for this connection. (i.e. un-ACKed bytes for this connection)

Since there is no direct way to determine the congestion window at the TCP sender(), the outstanding unacknowledged data is used to estimate the congestion window.

The red line shows window size advertised by the receiver.

The blue line show bytes in flight for the user.

Since $\text{bytes_in_flight} \leq \min(\text{congestion_window}, \text{receiver_window})$

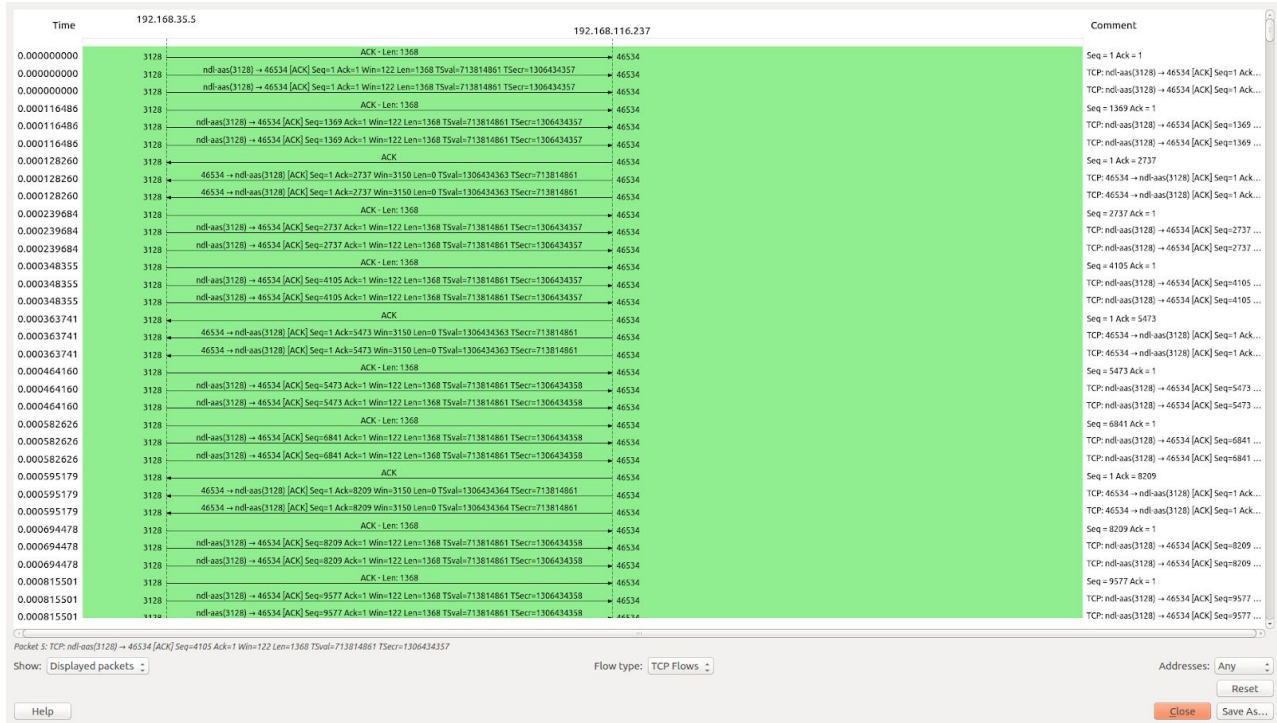
And we observe blue line is always below red line, we can say with some assumptions that blue line plots congestion window

(Though there is a possibility that `bytes_in_flight` is less due to receiver sending ACK before the `receive_window` was full. Hence this graph gives a rough estimate of `congestion_window` (with the assumption that the above condition is not happening much)).

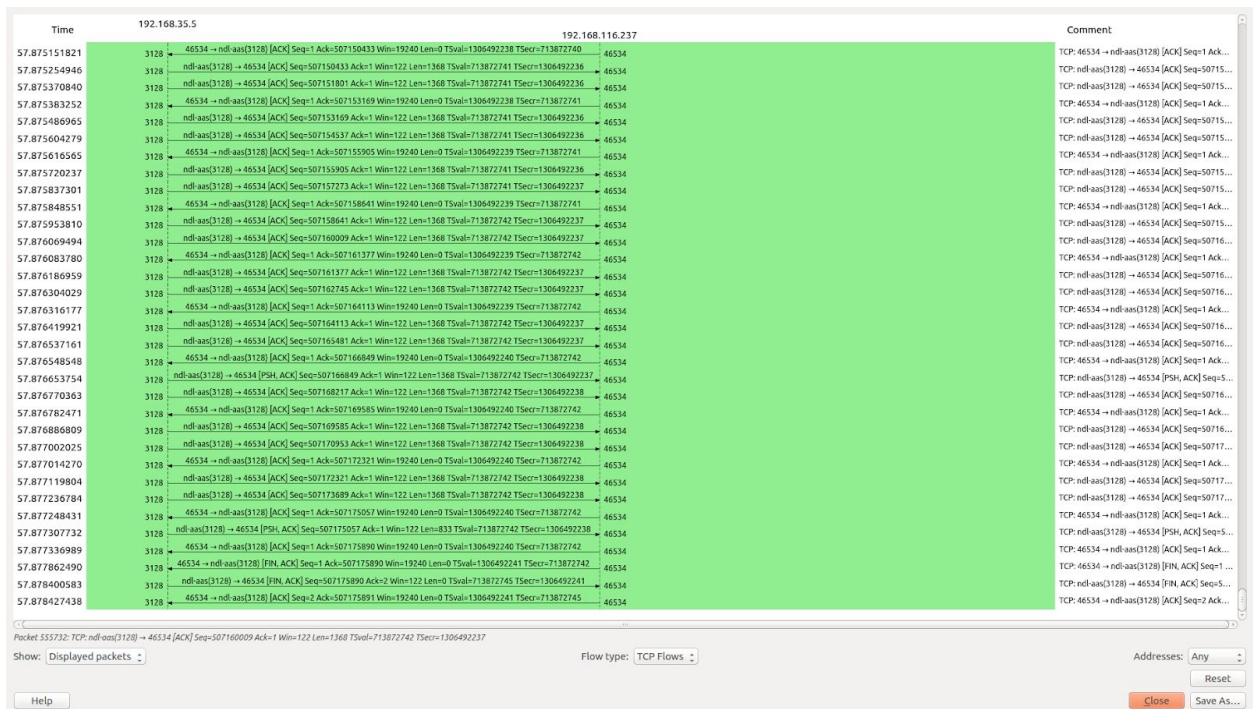
We observe that there is a log of fluctuation in the graph (thereby showing an unstable network (instability shown by above RTT vs time graph also))



c) The flow graph feature shows a sender and a receiver view of the packet flow. (listing all seq-nos. & ack-nos. , etc). This is a really big graph as it tells of entire conversation. I plotted TCP flow for the conversation between intranet and my IP. The first screenshot is flow-graph first page. The initial seq,ack no. is 1 showing tcp handshake is starting (numbers are relative). And also sender is sending many pay-load (therefore this is not stop & wait). For first few iterations sender is sending same packet twice(meaning timeout must have occurred). Also receiver sends 2 acks with seq=2737(1369+1368), meaning dup-ack. So on the conversation progresses.



The second screenshot is flow graph last page. The conversation ends with sender sending FIN ACKS to receiver (asking to close connection). Sender sends ACK to acknowledge that connection close message has been received.

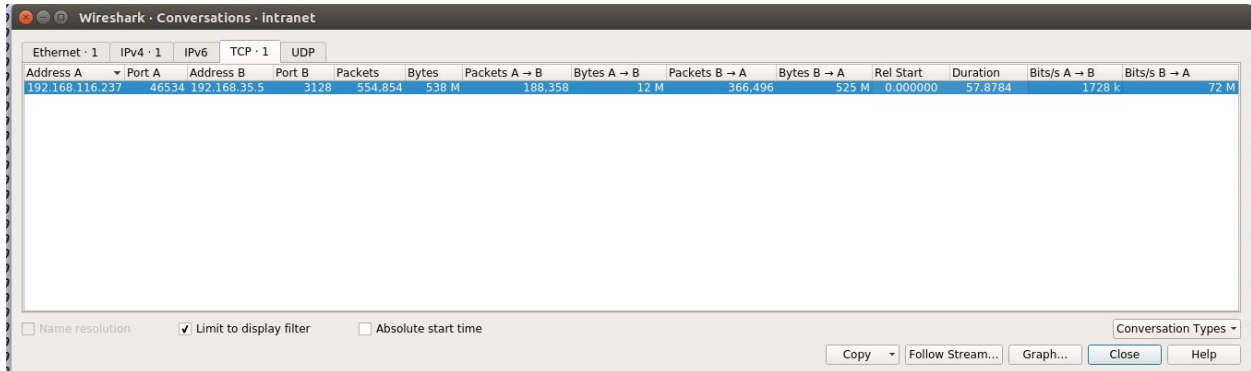


d)

Average Throughput from sender -> receiver :: 72 Mb/sec

Average Throughput from receiver -> sender :: 1728 kb/sec

Sender -> receiver is higher because here packets with payload is being sent, whereas in receiver -> sender just ACKs are being sent.



e)

The graph shows The size of the advertised receiver window incl. scaling factor.

The green plot shows receiver_window. In the plot

0-1s size is constant at 3510B

At 1s value increases to 6300B

1-31 s value ~6300B with fluctuations in middle

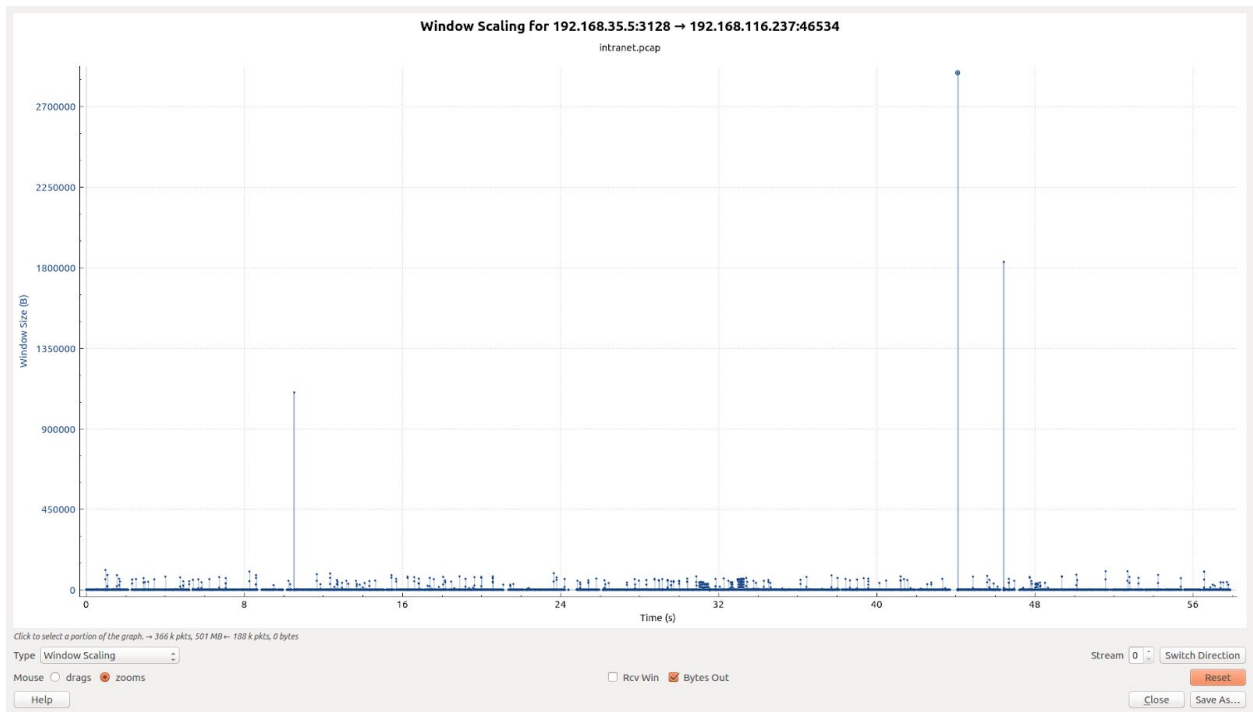
31-32s value decreases to 3450B

At 32s value shoots up to 19200B(with a few fluctuations)

The reason for the sudden shooting up can be that the initially sender sends smaller number of packets so as not to overwhelm the receiver. When sender sees it can send more packets, the value shoots up for increase throughput.

The 2nd graph is of Outstanding bytes vs time

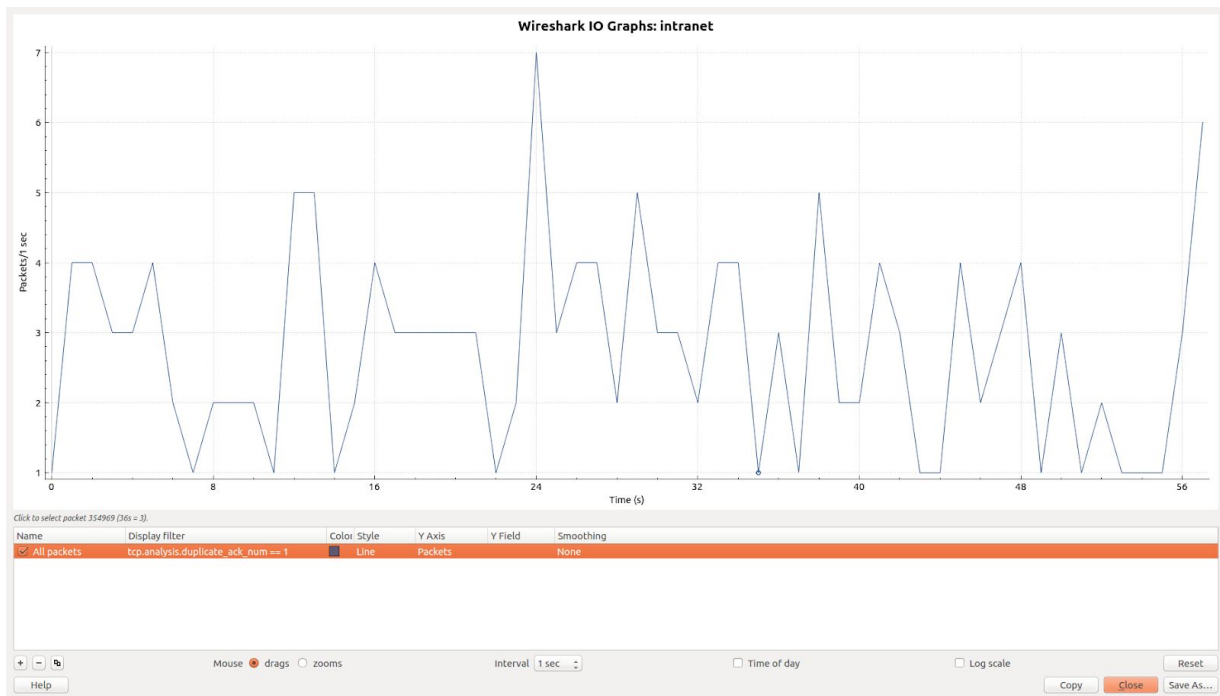
I.e bytes that are unacked at receiver side. Most time it is between 1500B - 3000B (mostly below receive_window but for sometimes there are some shoot up beyond receive_window too)



f) 1-duplicate Ack

Fraction :: 160 / 555772 (0.0%)

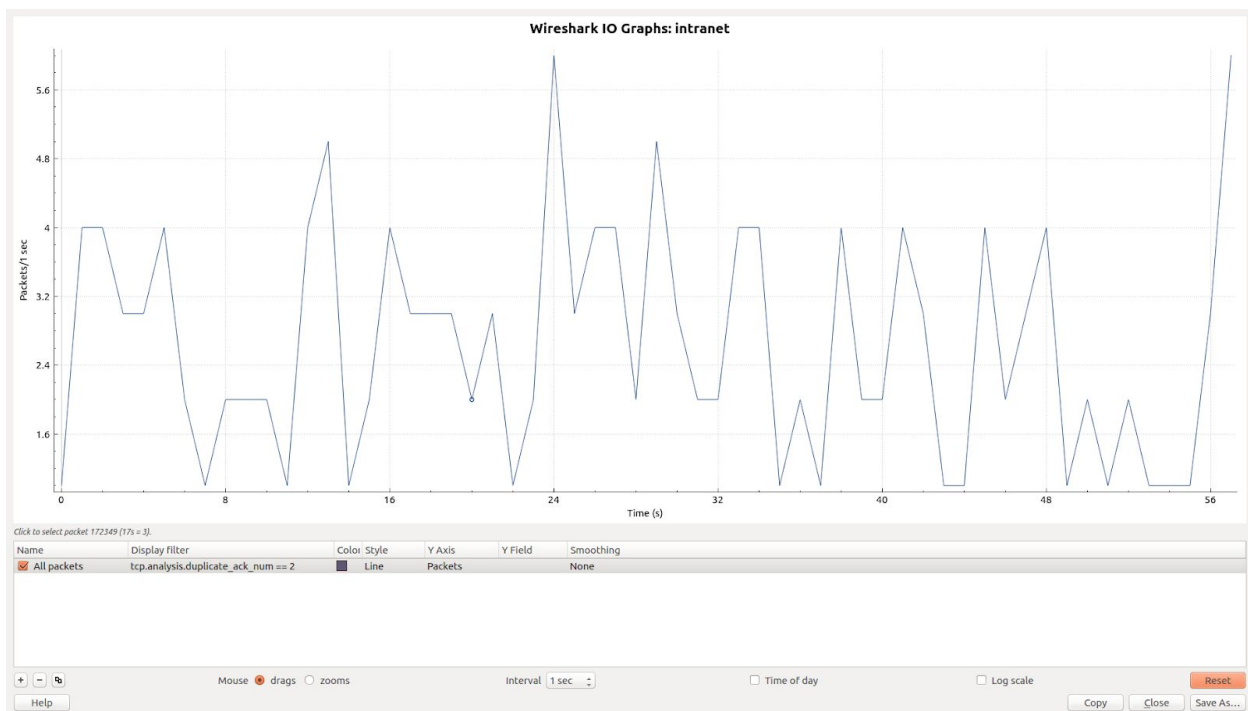
Command used :: **tcp.analysis.duplicate_ack_num == 1**



2-duplicate Ack

Fraction :: 153 / 555772 (0.0%)

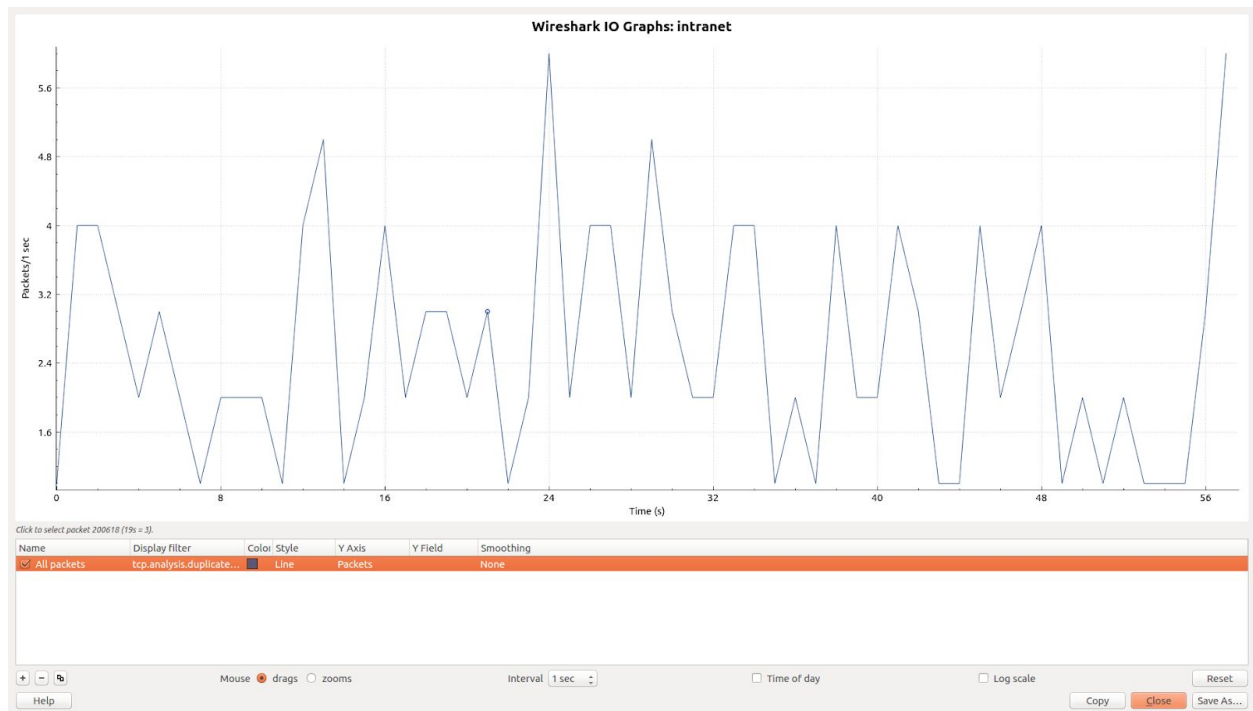
Command used :: **tcp.analysis.duplicate_ack_num == 2**



3-duplicate Ack

Fraction :: 149 / 555772 (0.0%)

Command used :: **tcp.analysis.duplicate_ack_num == 3**



Number of dup-Acks is less compared to other packets. There is difference in the above 3 graphs because for some lost packets it is not needed to go upto 3-dup ACK. 3-dup ACK is mostly used in congestion control (via fast retransmit).

Drive Links for PCAP files

intranet.pcap (for intranet capture)

<https://drive.google.com/open?id=1YJnjXx6cBlhVjDtB3ecMyJZfjLmDqR3C>