Exercise 1: Understanding the Impact of Network Dynamics on Routing

Question 1. Which nodes communicate with which other nodes? Which route do the packets follow? Does it change over time?

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Communication: Node 0: 1
Node 1: 0, 2, 4
Node 2: 1, 3
Node 3: 2, 5
Node 4: 1, 5
Node 5: 3, 4
Route: Node 0 \rightarrow \text{node } 1 \rightarrow \text{node } 4 \rightarrow \text{node } 5
Node 2 \rightarrow \text{node } 3 \rightarrow \text{node } 5
The routes do not change over time.
```

Question 2: What happens at time 1.0 and at time 1.2? Does the route between the communicating nodes change as a result of that?

time 1.0: link 1 - 4 is link-down and the route doesn't change. Packets can't transmit from node 0 to node 5.

time 1.2: link 1 - 4 is link-up and the route doesn't change. Packets waiting at node 1 go to node 4 and then node 5.

Question 3: Did you observe any additional traffic as compared to Step 3 above? How does the network react to the changes that take place at time 1.0 and time 1.2 now?

time 1.0, link 1 - 4 is link-down. The routing protocol discovered a different route (0 \rightarrow 1 \rightarrow 2 \rightarrow 3 \rightarrow 5) and that is used to transmit the packets.

time 1.2, link 1 - 4 is link-up. Since the cost of original route $(0 \to 1 \to 4 \to 5)$ is lower than the discovered route $(0 \to 1 \to 2 \to 3 \to 5)$, the routing protocol used the original route.

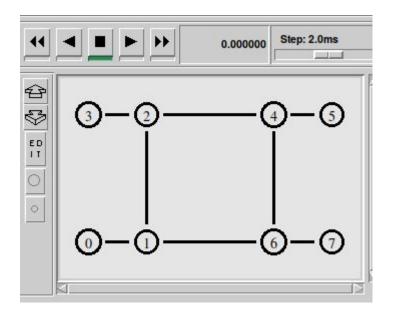
Question 4: How does this change affect the routing? Explain why.

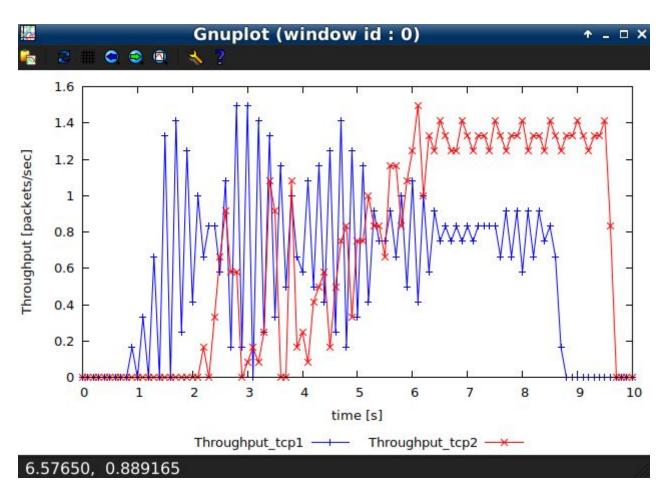
This change means the cost of link 1-4 is 3. Hence, the route is $0 \to 1 \to 2 \to 3 \to 5$ as it's lower than the cost of link $0 \to 1 \to 4 \to 5$.

Question 5: Describe what happens and deduce the effect of the line you just uncommented.

It changes the cost of link 1 - 4 to 2. The route 0 -- 1 -- 4 -- 5 and route 0 -- 1 -- 2 -- 3 -- 5 are having equal cost now. Therefore, the traffic would be separated to both routes when the flow reaches node 1.

Exercise 2: Setting up NS2 simulation for measuring TCP throughput





Exercise 3: Understanding IP Fragmentation

Question 1: Which data size has caused fragmentation and why? Which host/router has fragmented the original datagram? How many fragments have been created when data size is specified as 2000?

The data size that caused fragmentation is 2000 and 3500. The screenshot shows the link between two routers runs a 1500 MTU. If the size of a packet exceeds the MSS or the MTU of TCP, the only way to send the packet is fragment it.

The IP fragmentation occurs at the IP layer. Not only the source host fragmented the original datagram, but also the router. Because the MTU vary in different networks. If the MTU of a certain network on the transmission path is smaller than the MTU of the source network, the router may fragment the IP datagram again. In here, only the source host (192.168.1.103) fragmented the original datagram.

2 fragments.

Question 2: Did the reply from the destination 8.8.8.8. for 3500-byte data size also get fragmented? Why and why not?

Yes. The destination host will reassemble the fragments to IP datagram and then send back the IP datagram. The size is 3500 byte which is bigger than the MTU of the same network. Therefore, the reply from the destination 8.8.8.8 needs to get fragmented.

Question 3: Give the ID, length, flag and offset values for all the fragments of the first packet sent by 192.168.1.103 with data size of 3500 bytes?

No	ID .	Length	Flag	Offset
1				0
2				185
3	0x7a7b	568	0x0172	370

Question 4: Has fragmentation of fragments occurred when data of size 3500 bytes has been used? Why and why not?

No. We can see that the packets have been fragmented to 3 fragments and none of these fragment has fragmentation occurred. From the picture below, two of the fragments size are 1480 bytes and the other one is 548 bytes. Their sizes are smaller than the MTU of the current network.

```
  [3 IPv4 Fragments (3508 bytes): #52(1480), #53(1480), #54(548)]
  [Frame: 52, payload: 0-1479 (1480 bytes)]
  [Frame: 53, payload: 1480-2959 (1480 bytes)]
  [Frame: 54, payload: 2960-3507 (548 bytes)]
```

Question 5: What will happen if for our example one fragment of the original datagram from 192.168.1.103 is lost?

When one fragment lost, the source host has to resend the entire IP datagram again. The reassembly of fragmented data only occurs at the IP layer of the destination host. Because the IP layer does not have a timeout retransmission mechanism, the TCP has to resend the entire TCP segment after the timeout when a piece of the TCP segment is lost. The segment

corresponds to a datagram.	an IP datagram	. There is no way	to retransmit only one	fragment in a