CSC458: Problem Set 2

Due on Sunday, Nov $4^{\rm th},\,2018$

Zhongtian Ouyang 1002341012

BGP

- a)

 There are three routes to P Provider Q's BGP could receive. P can be arrived from Q through links 1, 2, 3
- From B to A, Provider P will choose to use link 1. From A to B, Provider Q will choose to use link 2.

Problem 2

IP

a)

For inbound traffic, all trafic need to be first send to the entry point IP address the organization holds, which corresponds to an specific address, the address of the headquarter, for example. And then send to the actual destination through the organization's own links. All the traffic enters the organization at a single point. This could be inefficient because the organization is international. The geographical address of the actual destination could be far away from the IP address's geographical address. There could be alternative shorter paths from source to destination.

- b) The organization can have its own routing table, which contains information about which destination ip correspond to which geographical location. When there is an outbound traffic, it could route the traffic to the exit point geographically closest to the destination.
- c) The organization should divide its internal network to several subnets base on geographical location. For example, the servers in USA share a subnet while servers in Asia share another. The routers in outside world could then add seperate routing entries for each subnet.
- d)
 The internal routers need to be able to recognize internal destinations and have entries to all the other internal networks using internal routes.

Problem 3

Sequence Number

When setting up an TCP connection, the ISN(initial sequence number) is set to an random number between 0 and $2^{32} - 1$ for reasons discussed in lecture. Therefore, if the ISN is a very large number close to $2^{32} - 1$, without wrap around, the sequence number runs out quickly after a few packets.

Fast retransmit

Suppose the segment S0 is sent at t = 0. With lossless connection, S7 is sent at t = 700; ACK for S0 should arrive at t = 800, so the packet S8 can be sent at t = 800.

Now with a loss of S0, we can't send S8 at t=800 because the window size is 8. We received the three duplicate ACKs at t=900, 1000, 1100. With fast retransmit, S0 is retransmitted at t=1100, ACK of S0 is received at t=1900. The packet S8 can be sent at t=1900 time lost t=1900 - t=1100 - t=11

Problem 5

TCP slow-start

Since the client hosts are connect directly to our routers, we can examine the headers of all the tcp packets. Then, we can determine the congestion window from the outbound tcp packet's sequence number and inbound tcp packet's ACK. And by observing the congestion window, we can determine whether the client is using slowstart.

For example, when the tcp connection first established, we should observe the sequence number is only one MSS more than the ACK number, aka the congestion window size is 1 MSS.

Also, after a segment with the same sequence number is retransmitted, we should observe the congestion window decreased at least by half. If the three duplicated ACKs is observed, the congestion window should be decreased by half. If not, then there is a timeout, the congestion window should be reduced to a very small number such as one.

Problem 6

Congestion

The only time the traffic will pass through R1 is when Hs on the left side of R1 communicate with Hs on the right side of R1. Since all links are full-duplex, we only need to consider traffic in one direction. And since the graph is symmetrical, left to right is the same as right to left. In the worst case, the maximum throughput from left side to the right side is 4 * 1 MBps = 4 MBps, which is the capacity of the links of R1. So R1 can not be congested.

For all other routers, the router would be congested if all its parent's bandwidth are going toward one of its children, for example, right child. It would be even worse if the left child also send traffic toward right child. The situation is shown in following graph. R3-B link need a bandwidth of 3x but only got a capacity of x.

$$\mathbb{R}_{2}$$
 \mathbb{R}_{2}
 \mathbb{R}_{3}
 \mathbb{R}_{3}
 \mathbb{R}_{3}
 \mathbb{R}_{3}

TCP Ports

- a) $A \rightarrow S$: Source Port: 12345 Destination Port: 23
- b) $B \to S$: Source Port: 23456 Destination Port: 23
- c) $S \to A$: Source Port: 23 Destination Port: 12345
- d) $S \rightarrow B$: Source Port: 23 Destination Port: 23456

e)

Yes. Because A and B are different hosts, the source port from A and source port from B are irrelavant.

f)

No. Since the are the same host, they have the same ip address. The post number must be different in order to identity difference processes and make sure the information is received by the correct process.

Problem 8

Checksum

Calculate 1's complement sums:

```
01010011 + 01100110 = 10111001, no over flow
```

10111001 + 01110100 = 100101101, 001011101 + 1 = 00101110

The checksum is the sum's 1's complement, which is 11010001.

We use one's complement of the sum as the checksum because by adding one's complement of the sum to the sum, we got all 1's. It is more efficient to check whether there is an 0 than comparing two numbers and determine whether they are the same.

The receiver add the three numbers and checksum together. If no error, the sum should be all 1's. If there is 0 in the sum, there is an error.

All 1-bit error will be detected, but 2-bit error can go undetected. For example, if the first two words become 01010010 and 01100111, the sum is unchanged.

Problem 9

AIMD congestion control

a.)

Since we increase 1 MSS everytime we received a batch of ACKs, with no loss, we need to receive 6 batches of ACKs to increase cwnd from 6 MSS to 12 MSS. We receive a batch of ACK every 1 RTT. So we need 6 RTT to increase cwnd from 6 MSS to 12 MSS.

```
b) Total MSS sent = 6+7+8+9+10+11=51 Average Throughput = 51 MSS / 6 RTT = 8.5 MSS/RTT
```

BGP

- a) Router 3c learns about the x from eBGP
- b) Router 3a learns about the x from iBGP
- c) Router 1c learns about the x from eBGP
- d) Router 1d learns about the x from iBGP

Problem 11

Dijkstra

Steps	S	D(t), p	D(u), p	D(v), p	D(w), p	D(y), p	D(z), p
0	{x}	inf, null	inf, null	3, x	6, x	6, x	8, x
1	$\{x, v\}$	7, v	6, v	3, x	6, x	6, x	8, x
2	$\{x, v, w\}$	7, v	6, v	3, x	6, x	6, x	8, x
3	$\{x, v, w, y\}$	7, v	6, v	3, x	6, x	6, x	8, x
4	$\{x, v, w, y, u\}$	7, v	6, v	3, x	6, x	6, x	8, x
5	$\{x, v, w, y, u, t\}$	7, v	6, v	3, x	6, x	6, x	8, x
6	$\{x, v, w, y, u, t, z\}$	7, v	6, v	3, x	6, x	6, x	8, x

Shortest paths:

 $x \to t : \mathbf{x}, \mathbf{v}, \mathbf{t}$

 $x \rightarrow u : x, v, u$

 $x \to v : \mathbf{x}, \mathbf{v}$

 $x \to w$: x, w

 $x \to y : x, y$

 $x \to z$: x, z

Problem 12

Sequence Number

a)

TCP sequence number has a range from 0 to $2^{32} - 1$. So there are 2^{32} possible sequence numbers. we increase sequence number by 1 every byte.

Maximum value of $L = 2^{32} = 4294967296$ bytes

b)

Number of segments = [4294967296/536] = [8012998.68657] = 8012999

For each segment, 66 bytes of various headers are added before sent out.

 $Total = 4294967296 + 8012999 \times 66 = 4823825230 \text{ bytes} = 38590601840 \text{ bits}$

 $38590601840/155000000 = 248.971624774s \approx 249s$