EECS 489 - FA 24 Discussion 6

TCP

Logistics

Due: Friday, Oct. 11 (right before Fall Break)

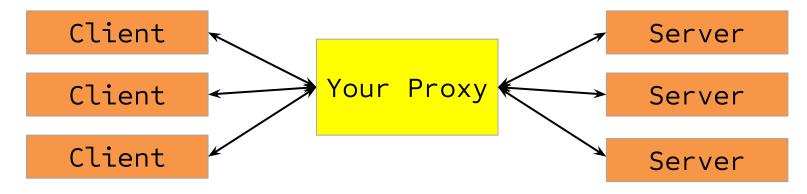
- Groups of 1-3 people
- 3 AG submits per day
- 3 late days across all 3 remaining projects
- Order of magnitude longer and more difficult than Assignment 1!

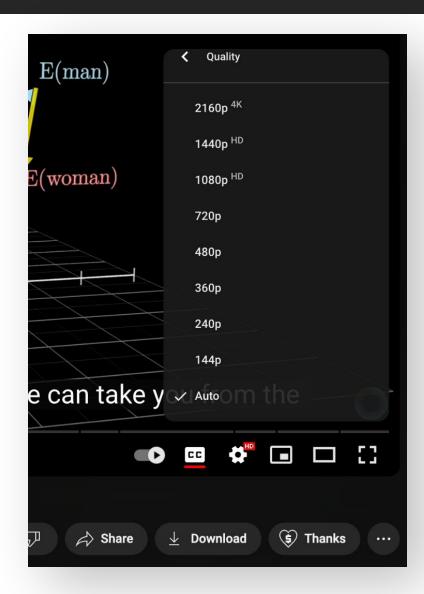
How to Develop:

• Develop on AWS like Project 1; you will have to set up a VNC Client on your computer, as we need a GUI for this project.

Part 1: HTTP Proxy for video streaming

- Sits between a client (web browser) and a standard video server.
- Modifies client requests to the video server to ask for the correct bitrate, based on constant measurement; echoes replies from the video server back to the client.
- Should be able to deal with multiple concurrent client connections!





Part 2: DNS Server with Load-Balancing

- Serves requests only for the URL of the video server
- Load-balances to different IPs using two strategies (configurable when the executable is called)
 - **Strategy 1:** Round robin
 - Strategy 2: Geographical proximity
- No hierarchical DNS lookup all queries go directly to this load-balancing server, which responds with an A-record

Integrating Part 1 and Part 2

- Have the HTTP proxy call the DNS server when a new client connects to figure out which server to route that client's traffic to.
- You can work on Part 1, Part 2 independently and then integrate at the end both are testable on their own.

TCP

Transmission Control Protocol (TCP) Header 20-60 bytes

source port number			destination port number
2 bytes			2 bytes
sequence number 4 bytes			
acknowledgement number 4 bytes			
data offset	reserved	control flags 9 bits	window size
4 bits	3 bits		2 bytes
checksum			urgent pointer
2 bytes			2 bytes
optional data 0-40 bytes			

- Provides a layer of **reliable transport**.
- Modeled as a **bytestream** rather than a stream of packets (even though this bytestream is divided into packets in practice).
- Provides **congestion control** mechanisms to dynamically discover the optimal speed for sending data.

- State variables maintained at sender:
 - Congestion Window Size (CWND): The actual window size is the minimum of CWND and the window size advertised by the receiver (RWND).
 - **Slow-start threshold (ssthresh):** The threshold at which the CWND update rate changes.
 - Timer: Standard sender-side timer for detecting timeout.
 - Duplicate ACK Count: How many duplicate ACKs have been received (0, 1, 2,
 3). This indicates a one-off packet drop.

- Events to update sender state:
 - ACK for new data
 - o **Timeout** on sender-side
 - Duplicate ACKs

- **ACK** for new data
 - Good! Increase CWND.
- **Timeout** on sender-side
 - Bad! Decrease CWND and recalibrate ssthresh.
- Duplicate ACKs
 - OK (implies one-off loss). Retransmit missing packet and decrease CWND slightly.

Consider transferring an enormous file of L bytes from host A to host B.

What is the maximum value of L such that we don't run out of TCP sequence numbers?

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2^32 Bytes

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Given $L = 2^32$ bytes, find how long it takes to transmit the file Assume:

- MSS (max segment size) = 1460 Bytes
- MTU (max transmission unit) = 1500 Bytes
- 128 Mbps link from A to B
- Ignore flow and congestion control, assume A sends as fast as possible contiguously.

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```
(2^32 / 1460 * 1500) / (128 * 10^6 / 8) = 276 s amount of data / bandwidth
```

Host A (sender) and B (receiver) are communicating over a TCP connection.

Assume the following events happen in order:

- B has received the first 127 bytes of the flow from A, this consumes seq num 0-126
- A then sends two segments, S1 (80 bytes of data), S2 (40 bytes of data)
- S1 has sequence num 127, source port 30302, destination port 80
- B sends ACK1 and ACK2 to A when it receives the first / second segment respectively

Assume S1 and S2 arrive in order.

For S2, what are the sequence num, source port and destination port?

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Destination Port: 80

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HTTP (Port 80).

Consider sending a large file over a lossless TCP connection. Assume:

- TCP uses AIMD for congestion control with slow start
- ssthres = 16 MSS (Maximum Segment Size)
- Approximately constant RTT
- CWND starts at 1 MSS
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CWND: 1 2 4 8 16 17 18 19 $20 \rightarrow 8$ RTT

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What is the average throughput (in terms of MSS and RTT) of the above process?

(1+2+4+8+16+17+18+19) / 8 = 85 MSS / 8 RTT

Thanks

Have a good one!