CS118 Project 2 Report

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1 Introduction

This report outlines the implementation of TCP and difficulties our group faced in this project. This project aims to deepen our understanding of the underlying protocols in TCP as well as how TCP deals with packet loss.

2 Implementation

We implement some functions of reliable data transfer protocol to using UCP in C and C++.

To compile the program: \$ make

To run the program:

Client side: \$./client <HOSTNAME-OR-IP> <PORT> <FILE_NAME>

Server side: \$./server <PORT>

2.1 Header format

We created a struct type called **header** to define a packet header. The struct header contains

- int16_t seqnum, sequence number
- int16_t acknum, ack number
- char flags, character sequence for ACK, SYN, and FIN flags. Since char type is 8 bits but we only need 3 bits for our flag(s), we leave the first 5 bits unused.
- char pad[7], an array of characters that are initialized to be all zeros. This provides padding to make a header 12 bytes.

To obtain a character sequence of flags, we assign macros to ACK, SYN, and FIN and OR them together.

2.2 Messages

Server and client log to stdout for every packet received or sent with the following format:

- RECV <Seqnum> <Acknum> <cwnd> <ssthresh> <ACK> <SYN> <FIN>
- SEND <Seqnum> <Acknum> <cwnd> <ssthresh> <ACK> <SYN> <FIN> <DUP>

The SEND message has an additional DUP bit which indicates if this packet is a result of retransmission.

2.3 Connection class

We create a Connection class which deals with handshake, receiving and transmitting packets, and handwave on the server side. The reason for this is to provide another level of abstraction which makes it easier for us to maintain connections in server. Each time a SYN packet arrives on the server side, the server creates a Connection object, proceeds with the handshake protocol and takes appropriate actions. Once the handwave is done, the server deletes this Connection object and moves on. In other words, the server program only manages connections and takes the data read by Connection and writes it into appropriate file.

2.4 Timeouts

There are two timeouts in this program.

2.4.1 10s timeout

The 10s timeout occurs when:

- the server establishes a connection but receives no data from the client after 10 seconds since the connection establishment
- the server establishes a connection but receives no data from the client after 10 seconds since the last successful data packet transmit
- client receives no packets from server for more than 10 seconds

When this timeout occurs, the relevant side exits the program with an exit code 1 and an error message Error: 10 seconds timeout.

2.4.2 RTO timeout

RTO is fixed at 0.5 seconds. The RTO timeout occurs when client has not received an appropriate ACK from server for more than 0.5 seconds. The timeout triggers packet retransmission and appropriate congestion control actions.

2.5 Timer

Timer is used in both Client and Connection in keeping track of the timeouts for each packet. Client has two timers: time10 for the 10s timeout and timer for the RTO timeout. Each Connection has time10 timer (and timer which is used only in handwave).

When a client sends a packet, client's time10 and timer is activated while the file has not been exhausted or the size of cwnd is non-zero. These two conditions are specified to avoid a situation where the timer is unnecessarily triggered when the file is already exhausted. Client's time10 is stopped when it receives any packet from server. Its timer is stopped when the corresponding ACK of the sent packet is received.

This follows the TA's suggestion to keep track of only one timer for RTO and constantly activate and deactivate it as a way of optimizing the program. Keeping track of one timer per each packet results in overhead which can reduce program's performance.

2.6 cwnd buffer

cwnd buffer is implemented by a vector of <struct header, string> where we store the header and body of a packet. Every time a client sends a packet, it inserts the packet into cwnd. Whenever an ACK is received, the client checks if the ACK is in order by comparing it against the first packet in cwnd. If yes, the relevant packet is removed from cwnd and slow start or congestion avoidance is performed. Otherwise, the number of duplicate ACKs is incremented.

2.7 Slow Start, Congestion Avoidance/ Slow Start

- Initially, cwnd is set to be 512 and ssthresh is set to be 5120.
- After handshake, the client starts sending packets in which the number of packets is controlled by cwnd.
- After each ACK is received, the client uses Slow Start if cwnd > ssthresh by incrementing cwnd by 1 packet (512 bytes) and Congestion Avoidance if cwnd < ssthresh by incrementing cwnd by 1/cwnd packet.
- A packet loss is detected when the number of duplicated ACKs is incremented to 3 or when the RTO timeout is reached.
- When packet loss occurs, client sets ssthresh = max(cwnd/2, 1024), set cwnd = 512, and retransmit data after the last ACK byte. Slow Start or Congestion Avoidance is performed after retransmission.
- cwnd is incremented by 1 packet every time additional duplicate ACK is received.
- Cumulative ACK is used during packet loss.

2.8 Server-side Buffering with rwnd

Instead of Go-Back-N with server side buffer as size 1, we choose to implement a full on buffer to reduce the number of retransmits on the client side. Theoretically speaking, the rwnd has size ∞ therefore window size is only bounded by cwnd. We maintain a vector/list of packets. Upon receiving a packet, we buffer it up. Then, for each element in the list of buffered packet

- 1. We check if the packet has the expected sequence number.
- 2. If yes, we return the data to Server to write to file, increment expected sequence number, remove the packet from the buffer, go back to step 1 for checking the rest of the array
- 3. If no, we simply buffer it up and does not return any data.

This seems like a dangerous algorithm, this can potentially goes up to $O(n^2)$ time if everytime we remove the last element in the list. We could have maintained a more complex data structure, such as a hash map that maps SEQ number to packet, or a list sorted according to SEQ number. However, since cwnd (which bounds the window size) is bounded by 20 packets, this operation is safely bounded. Therefore, a simpler algorithm helps since managing other states is already difficult enough. There is an hidden bug in this approach. We will get into that in section 3.1.

3 Difficulties

3.1 Problem of Lost ACK

We imagine that cwnd has 5 packets: [1, 2, 3, 4, 5]. Say that server has received and buffered 4 of them: [2, 3, 4, 5]. It should send 4 duplicate ACK1 to client. And client should retransmit packet 1 upon 3rd duplicate ACK. If server receives it, it will send back ACK6 and buffer will be emptied. Imagine that ACK6 is lost. Then, client will retransmit packet 1 upon RTO timeout. However, server will buffer it up also causing packet 1 to stay in the buffer forever, until the ACK number eventually wraps around. Now, we have 2 packet 1's in the buffer and of course our program will be confused. Initially, we think that there is no way around this situation. Since we cannot directly compare the sequence number in the s1 < s2 manner (sequence number can wrap around), this problem has to be left unsolved.

Solution: We realize that there is a maximum cwnd limit, which is 10240. The maximum sequence number is 25600, which is sufficient for us to determine if sequence number comes before or after. We wrote a helper function in the following manner.

```
bool is_s1LTs2(int16_t s1, int16_t s2) {
   if (s1 < s2) return true;</pre>
```

```
if (s2 < 10240) {
        int temp = 25600 - 10240 + s2;
        return s1 > temp && s1 <= 25600;
}
return false;
}</pre>
```

Now we can choose to ignore sequence numbers. Therefore, our implementation can work for large files with loss.

3.2 Problem of Frozen Timer

Imagine a scenario when the file has been exhausted, but cwnd is not emptied yet because we have not received the ACK for the sent packet. When an ACK is received, we will disable the timer for RTO. However, at this point, we no longer have packet to add to cwnd. Therefore, timer for RTO is never activated, and no packet ever gets transmitted. The states of both the server and the client are now frozen. Eventually, the 10s timeout arrives and kill the connection. This is bad for reliable transmission.

Solution: We change the RTO timer start condition. The timer should be activiated whenever there is still data in the cwnd buffer. Then, this solves the above mentioned error.