${\rm CMPEN362-Midterm\ Exam,\ Fall\ 2025}$

Name: (all capital letters) $\underline{\text{solution}}$

Student email:

Section: (circle one) Section 1 Section 2

P1	/10
P2	/10
P3	/10
Total	/30

Problem 1

Check all the correct answers. [1 pt per question]

1.	From a service view, the Internet is: □ a network of interconnected ISP networks □ communication links and packet switches ✓ an infrastructure that provides communication services to applications ✓ an infrastructure that provides programming interface to application developers
2.	The two key network-core functions are: ✓ determine the source-destination route □ guarantee delivery along the route ✓ forward packets from a router's input port to an output port according to the route □ ensure minimum throughput along the route
3.	The four sources of packet delay at a single hop include: queueing delay, propagation delay, ✓ transmission delay □ pipelining delay ✓ processing delay □ decryption and re-encryption delay
4.	What can NOT cause packet loss: □ buffer overflow at packet switch □ failure to pass checksum ✓ undetected bit error □ malicious attack
5.	The three tiers of DNS servers are: root DNS server, □ Tier-1 DNS servers ✓ authorative DNS servers ✓ TLD DNS servers □ local DNS servers
6.	The communication endpoints that send/receive messages over the Internet are \Box users \checkmark processes \Box hosts \Box sockets
7.	The header fields identifying a TCP socket are ✓ source IP address ✓ destination IP address ✓ destination port number
8.	The reasons for preferring TCP over UDP include: □ smaller header ✓ lossless data transfer □ smaller delay ✓ in-order delivery

9. A process running a P2P application is
\square always a client process
\checkmark a client process when initiating communication
\square always a server process
\checkmark a server process when waiting to be contacted
10. Suppose host A sends a TCP segment to host B with sequence number 212, acknowledgement number 70, and a payload of 20 bytes. The return segment from B to A will have acknowledgement number □ 212 ✓ 232 □ 70 □ 282

Problem 2

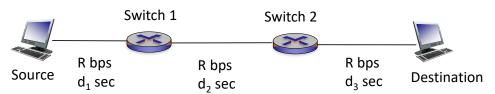


Figure 1: Problem 2.a illustration.

- a) Consider a source-destination pair connected by 2 packet switches via 3 links as in Fig. 1. Suppose link i (i = 1, 2, 3) has a bandwidth of R bps and a propagation delay of d_i seconds. Ignore queuing and processing delays. Suppose that an M-bit message is segmented into P packets of equal length.
 - (i) How long does it take for the first packet to arrive at the destination? [1 pt]

$$\sum_{i=1}^{3} \left(\frac{M}{PR} + d_i \right) = \frac{3M}{PR} + \sum_{i=1}^{3} d_i$$

(ii) How long does it take for all the packets to arrive at the destination? [1 pt]

$$\frac{3M}{PR} + \sum_{i=1}^{3} d_i + (P-1)\frac{M}{PR} = \frac{(P+2)M}{PR} + \sum_{i=1}^{3} d_i$$

(iii) How long does it take for all the packets to arrive at the destination if each packet has an H-bit header? [1 pt]

$$\frac{3(M/P+H)}{R} + \sum_{i=1}^{3} d_i + \frac{(P-1)(M/P+H)}{R} = \frac{(P+2)(M/P+H)}{R} + \sum_{i=1}^{3} d_i$$

(iv) What is the minimum message size M such that the last packet does not depart the source before the first packet arrives at the first switch (including headers)? [2 pt]

Last packet departs the source at $P \cdot \frac{M/P+H}{R}$ First packet arrives at Switch 1 at $\frac{M/P+H}{R} + d_1$ Minimum M is given by: $P \cdot \frac{M/P+H}{R} \geq \frac{M/P+H}{R} + d_1$ [1 pt] $\implies M \geq \frac{d_1PR}{P-1} - PH$ [1 pt]

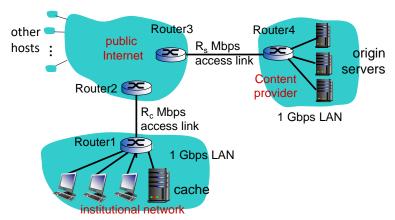


Figure 2: Problem 2.b illustration.

- b) Consider serving web objects from origin servers to hosts as in Fig. 2. Suppose that each object is 1M bits, hosts in the institutional network generate 10 requests/sec, and other hosts generate 100 requests/sec. The average delay to send a request from Router2 to Router3 and obtain response is 1 second. Ignore propagation delays for access and LAN links. Let $R_c = 11$ Mbps and $R_s = 120$ Mbps. Answer to precision 0.001.
 - (i) What is the delay for a host in the institutional network to obtain one object from origin server? [2 pt]

 Total delay = server LAN delay + server access delay + Internet delay + client access delay + client LAN delay [0.5 pt] server/client LAN delay = 1M bits/1 Gbps = 1 ms server access delay = $\frac{1}{120-100-10}$ = 0.1 sec [0.5 pt] client access delay = $\frac{1}{11-10}$ = 1 sec [0.5 pt] \Rightarrow total delay = 0.001 + 0.1 + 1 + 1 + 0.001 = 2.102 sec [0.5 pt]
 - (ii) Now deploy a web cache in the institutional network with hit rate 0.5. What is the average object downloading delay for hosts in the institutional network? [3 pt]

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Hit delay = LAN delay = 1 ms [0.5 pt]

Miss delay = server LAN delay + new server access delay + Internet

delay + new client access delay + client LAN delay [0.5 pt]

new server access delay = \frac{1}{120-100-5} \approx 0.067 sec [0.5 pt]

new client access delay = \frac{1}{11-5} \approx 0.167 sec [0.5 pt]

miss delay \approx 1.236 sec (also correct: 1.235 sec) [0.5 pt]

\Rightarrow average delay = 0.5· hit_delay + 0.5· miss_delay \approx 0.619 sec

(also correct: 0.618 sec) [0.5 pt]
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Problem 3

Host S wants to send segments alternatingly to Hosts A and B. Each of A and B is connected to S via a dedicated channel that can lose/corrupt (but not reorder) segments. Design a stop-and-wait transport protocol to make sure that the receivers receive segments in the order of: A, B, A, B...



Figure 3: Problem 3 illustration.

You can use the following procedure calls:

- rdt_send(data): called by upper layer to send data in 'data';
- rdt_rcv(rcvpkt): called by lower layer after receiving packet 'rcvpkt';
- from_host(rcvpkt, hostid): true if packet 'rcvpkt' is from host 'hostid';
- has_seq(rcvpkt, seqnum): true if packet 'rcvpkt' has sequence number 'seqnum';
- corrupt(rcvpkt): true if packet 'rcvpkt' is corrupted;
- udt_send(sndpkt, hostid): call lower layer to send packet 'sndpkt' to host 'hostid';
- extract(rcvpkt, data): extract payload of packet 'rcvpkt' into data structure 'data';
- deliver(data): call upper layer to deliver data stored in 'data';
- make_pkt(seqnum, data), make_pkt(seqnum, ACK): return a data or acknowledgement packet with sequence number 'seqnum';
- start_timer: start timer;
- stop_timer: stop timer;
- timeout: called when timer runs out.

In addition, use "!" for negation, "&&" for logical AND, and "||" for logical OR.

a) For the FSM at the sender S as shown in Fig. 4, give the content of states 5, 6, 7, and 8 following the convention of the given states [1 pt], and describe the meaning of each state. [1 pt]

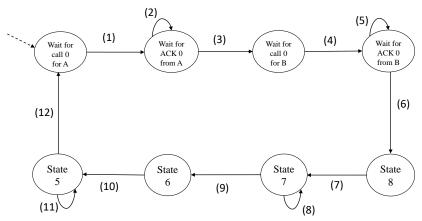


Figure 4: Problem 3: sender FSM.

state 5: "wait for ACK 1 from B", meaning waiting for acknowledgement with sequence number 1 from B

state 6: "wait for call 1 for B", meaning waiting for application data to be sent with sequence number 1 to B

state 7: "wait for ACK 1 from A", meaning waiting for acknowledgement with sequence number 1 from A

state 8: "wait for call 1 for A", meaning waiting for application data to be sent with sequence number 1 to A

Note: -0.5 pt if the order of states is reversed

- b) Complete the event-action list for the following transition links. [4 pt]

 - (2) timeout
 udt_send(sndpkt, A)
 start_timer
 - (3) rdt_rcv(rcvpkt) && !corrupt(rcvpkt) && from_host(rcvpkt, A) && has_seq(rcvpkt, 0) stop_timer

c) Can we use the receiver FSM of one of the protocols learned in class for A? If so, give the protocol name [1 pt], its states [1.5 pt], and the meaning of each state [1.5 pt].

Yes, we can use rdt3.0 (or rdt2.2) receiver. [1 pt]

States and meaning:

"wait for 0 from below", meaning waiting for packet with sequence number 0 from the sender (that will be delivered through the network layer below) [1.5 pt]

"wait for 1 from below", meaning waiting for packet with sequence number 1 from the sender (that will be delivered through the network layer below) [1.5 pt]

Complete FSM (not required):

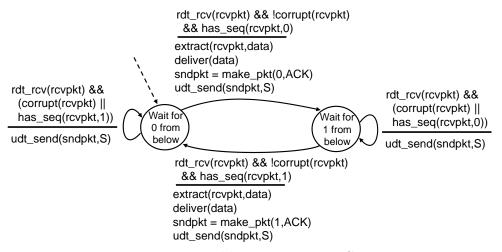


Figure 5: Problem 3: receiver FSM.