COMP-4320 Final Exam

Due Aug 3 at 10:30am

Points 100

Questions 50

Available Aug 3 at 8am - Aug 3 at 10:30am 2 hours and 30 minutes

Time Limit 150 Minutes

Instructions

There are 50 questions in this Final Exam.

This is a CLOSED note exam. Using your e-book, notes or textbook is NOT ALLOWED. **However, you are allowed two pages of blank scratch papers.** You may use two sheets of scratch paper. **You are also allowed to use a scientific calculator.** You must remain in front of your computer for the duration of the exam. NO BATHROOM BREAKS. Cell phones, tablets, laptops, smart watches, and any other electronic devices are NOT PERMITTED. Failing to follow these instructions could result in a violation.

This quiz is no longer available as the course has been concluded.

Attempt History

	Attempt	Time	Score
LATEST	Attempt 1	40 minutes	92 out of 100

(!) Correct answers are hidden.

Score for this quiz: 92 out of 100

Submitted Aug 3 at 8:42am This attempt took 40 minutes.

Question 1

1 / 1 pts

Consider a circuit-switched network shown in Figure 1, showing four channels/circuits in each of the links. There may be many hosts connected to each router.

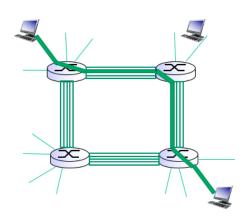


Figure 1.

The maximum number	of simultaneous	connections	that can	be in
progress at any time is	16.			

- True
- False

False

HTTP response messages may contain an empty message body. True

Question 3 2 / 2 pts

In the Selective Repeat protocol, the sender window size and the receiver window size must be the same.

○ True	
False	
Question 4	2 / 2 pts
Two-way handshakes can be used to es	stablish TCP connections.
○ True	
False	
Question 5	0 / 1 pts
Internet Control Message Protocol (ICM	P) is mainly used for routing.
True	
○ False	
Question 6	1 / 1 pts
Internet Protocol (IP) does not provide r	eliable delivery service for data.
True	

Incorrect

False			

Question 7	1 / 1 pts
Maximum Transfer Unit (MTU) is a standardized value for all ne links.	etwork
O True	
False	

Network Address Translation (NAT) is method of remapping an IP address to a machine's physical address. True False

Question 9	1 / 1 pts
Routing algorithm determines end-to-end path through network.	
True	

Fal	se

Incorrect	Question 10	0	/ 1	pts

Dynamic Host Configuration Protocol (DHCP) allows host to dynamically obtain its MAC address from a server when it joins a network.

True

False

Distance Vector mainly uses Bellman-Ford algorithm. True False

Question 12 2 / 2 pts

Suppose there are N routers from source to destination. Let L denotes the number of bits in a packet, R denotes the transmission rate. What is the total end-to-end delay in sending one packet from source to destination?

$\bigcirc \frac{2N \times L}{R}$
$\bigcirc \frac{L}{R}$
\bigcirc N
\bigcirc $\frac{N \times L}{R}$

Answer Questions 13 and 14 using the following information. Suppose there are in total 15 users sharing a 200 Mbps link. Also suppose each user requires a bandwidth of 25 Mbps when transmitting, but each user transmits only 30 percent of the time.

Question 13	2 / 2 pts
Suppose circuit switching is used, what is the maximum nurusers that can be supported?	mber of
8	
○ 13	
O 6	
O 15	

Suppose **packet switching** is used, what is the probability that exactly one user is transmitting?

- 0.002
- 0.3
- 0.7
- 0.031

Question 15

2 / 2 pts

Suppose there is exactly one packet switch between a sending host and a receiving host. The transmission rates between the sending host and the switch and between the switch and the receiving host are R1 and R2, respectively. Assuming that the switch uses store-and-forward packet switching, what is the total end-to-end delay to send a packet of length L? (Ignore queuing, propagation delay, and processing delay.)

- $\frac{L}{R1+R2}$

- $\frac{L}{R2}$

Again, suppose as in the previous question that there is exactly one packet switch between a sending host and a receiving host. The transmission rates between the sending host and the switch and between the switch and the receiving host are both *R*. Assuming that the switch uses store-and-forward packet switching, what is the total end-to-end delay to send **three** packets of length L each? (Ignore queuing, propagation delay, and processing delay.)

- $\frac{6L}{R}$
- $\frac{3L}{R}$
- $\frac{3L}{2R}$
- $\frac{4I}{R}$

Answer Questions 17 to 19 using the following information:

Consider the scenario shown below, where there are 3 connections between the client and the server. Each connection consists of 5 links with transmission rates as shown in Figure 2. All connections use a shared link R, whose capacity of 120 Mbps is shared equally among the 3 connections.

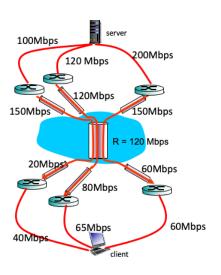


Figure 2.

Question 17 2 / 2 pts

What is the maximum achievable end-to-end throughput for the client and the server if only one connection can be used?

- 120 Mbps
- 200 Mbps
- 40 Mbps
- 20 Mbps

Question 18 2 / 2 pts

If the server can use all 3 parallel connections to send data, what is the maximum throughput that the server can achieve?

○ 715 Mbps		
100 Mbps		
350 Mbps		
140 Mbps		

Assuming that the server is sending at the maximum rate possible, what is the link utilization for the shared link R? 48% 100% 83% 50%

Incorrect Question 20 0 / 3 pts

The following are the Internetworking principles that are used for successfully developing the Internet:

- i. Stateless routers
- ii. Reliable connections
- iii. Decentralized control
- iv. Minimalism and autonomy

All of the above		
ii, iii and iv only		
i, ii and iv only		
i, iii and iv only		

What is the main reason for using layering principle for organizing the different modules of a complex network system? Each layer is autonomous It prevents deadlocks It enables minimal interaction between the modules It makes network access more efficient

Question 22 2 / 2 pts

To simplify development of complex network systems, we break the system into simpler modules, where each module consist of two parts: (1) interface which is well-known and accessible by users, and (2) implementation which is hidden from users. What is the main purpose for hiding the implementation?

	at users will no					
Prevents users	information in	the impleme	entation fro	m being re	vealed to th	e
O It prev	vents depende	ncy and dea	ıdlock			
The imple	ementation can	ı be modified	d without n	nodifying th	e user prog	rams

Why are most network applications in the Internet based on stateless client-server model, e.g. stateless HTTP protocol? There are other more efficient reliability algorithms It is inefficient to maintaining state information and store them in persistent storage It is difficult to maintain states of the server and client that are consistent with each other State information can be lost or corrupted

Answer Questions 24 and 25 using the following information.

Consider Figure 3 that shows the transmission of data segments and ACKs in TCP.

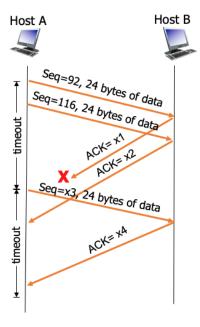
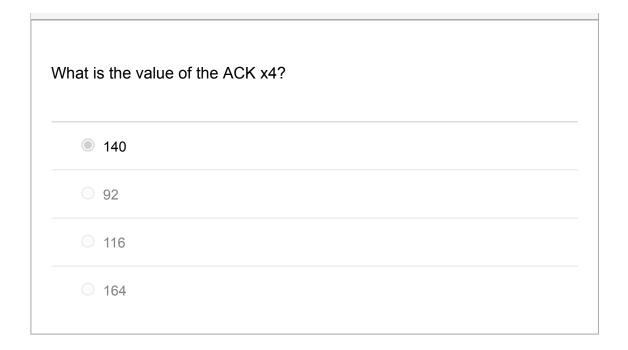


Figure 3.

Question 24	3 / 3 pts
What is the value of the Sequence Number x3?	
92	
O 164	
O 116	
O 140	

Question 25 3 / 3 pts



Answer Questions 26 to 27 using the following information.

Consider Figure 4 that shows the transmission of data segments and ACKs in TCP.

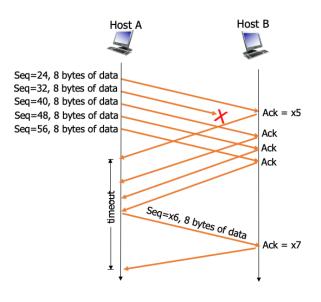


Figure 4.

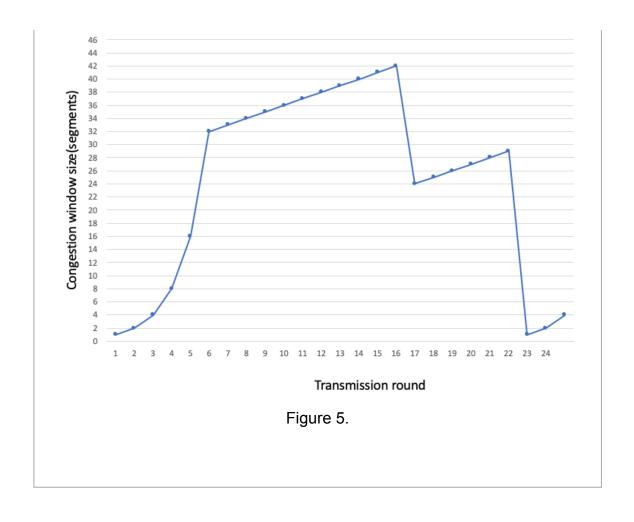
Question 26 2 / 2 pts

What is the value of the Sequence Number x6?	
O 56	
O 40	
32	
O 64	

Question 27	2 / 2 pts
What is the value of the ACK x7?	
O 56	
O 32	
O 40	
64	

Answer Questions 28 and 29 using the following information.

Figure 5 shows a behavior of TCP Reno congestion control.



Question 28	2 / 2 pts
What would be the most possible event that happen at 16 th tran	nsmission?
Triple duplicate ACKs received.	
The threshold is maxed out.	
The connection is broken.	
There is a timeout.	

Question 29 2 / 2 pts

hat would be the most possible event hap	ppen at 22 th transmission?
There is a timeout.	
The connection is broken.	
Triple duplicate ACKs received.	
The threshold is maxed out.	

Answer Questions 30 to 32 using the following information.

Consider the following institutional network that is connected to the Internet as shown in Figure 6.

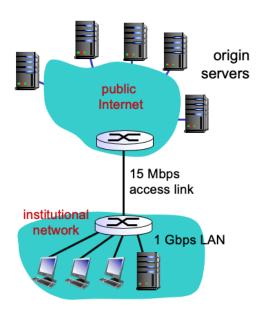


Figure 6.

Suppose that the average object size is 450,000 bits and that the average request rate from the institution's browsers to the origin servers is 32 requests per second. Also suppose that the amount of time it takes from when the router on the Internet side of the access link forwards an HTTP request until it receives the response is 3 seconds on average. Model the total average response time as the sum of the average access delay (that is, the delay from Internet router to institution router) and the average

Internet delay. Assume that if the utilization at the access link is less than 60%, then the queuing delay is 0. (Ignore the propagation delay and processing delay at the access link.) Answer the following three questions.

Question 30	2 / 2 pts
What is the access link utilization?	
O 60%	
O 99%	
○ 30%	
96%	

Question 31	3 / 3 pts
What is the total average response time?	
3.75 sec	
○ 1.8 sec	
minutes	
○ 3 sec	

Question 32	3 / 3 pts
Now suppose a cache is installed in the institutional LAN. Suppose time.	oose the
minutes	
○ 3.75 sec	
3 sec	

Question 33	3 / 3 pts
TCP uses a variation of the following reliable data transfer proto	ocol.
Go Back N protocol	
Alternating Bit protocol	
Selective Repeat protocol	
Stop and Wait protocol	

Question 34 3 / 3 pts

Why does TCP use the variation of the protocol in Question 33 above?

Because it requires fewest retransmission after a timeout.

Because it recovers quickly after triple duplicate ACKs.

Because each unacked segment has a separate timeout timer.

Because the implementation is simpler and most networks are reliable.

Question 35 3 / 3 pts

In the Selective Repeat protocol, what is the relationship between the sequence number size S and the sender window size W?

- S >= 2 X W + 1
- S >= W + 1
- S >= W
- S >= 2 X W

Question 36 3 / 3 pts

In the Go Back N protocol, what is the relationship between the sequence number size S and the sender window size W?

S >= W + 1

○ S >= 2 X W	
○ S >= W	
S >= 2 X W + 1	

Question 37 3 pts

To determine the appropriate timeout value to use for TCP, a sender X must estimate the round trip time (RTT) by sampling the RTT. Suppose, to compute this sample RTT, X sends a segment S at time t0 to Y but X did not receive the ACK before it times out and retransmits S at time t1. It then receives an ACK from Y at time t2. Suppose, X then computes the sample RTT as t2 - t1. Why is the sample RTT as computed incorrect?

- Because the sample RTT should be computed as t2 t0.
- Because the ACK may be for a different segment transmission.
- Because the ACK may be for the first S transmission.
- Because the sample RTT may have some variation.

Question 38	3 / 3 pts
How does TCP handle the above scenario in Question 37?	
It will ignore the sample RTT for this segment.	

It will calculate the average estimated RTT to correct the problem

It will	determine	if the	ACK is	s for	the	correct	S	transmission.	

It will add the safety margin for the variation in RTT.

Question 39 3 / 3 pts

Suppose the Autonomous System AS1 learns from inter-AS protocol that Subnet *X* is reachable from AS3 and from AS2 (see Figure 7 below). Which is the next router that Router 1d will forward packets with Destination X?

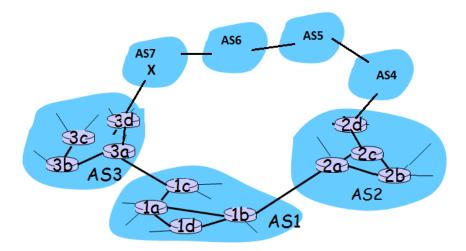


Figure 7. Hierarchical Routing in Autonomous Systems with Subnet X

○ 1a			
○ 1c			
1b			

None of these listed.

Answer Questions 40 to 43 using the following information.

Consider a Network Address Translation (NAT) router as shown in Figure 8.

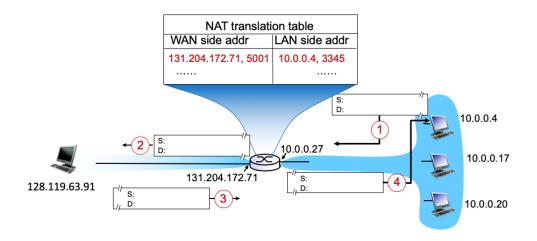


Figure 8

Three hosts, with private IP addresses 10.0.0.4, 10.0.0.17, 10.0.0.20, respectively, are in a local network behind a NAT router that sits between these three hosts and the Internet. IP datagrams being sent from, or destined to, these three hosts must pass through this NAT router. The router's interface on the LAN side has IP address 10.0.0.27, while the router's address on the Internet side has IP address 131.204.172.71.

Suppose that the host with IP address 10.0.0.4 sends an IP datagram destined to host 128.119.68.91. The source port is 3345 and the destination port is 80.

Question 40 1 / 1 pts

At Step 1, what are the source IP address, destination IP address and port numbers?

Source IP: 10.0.0.27, 80

Destination IP: 131.204.172.71, 5001

Source IP: 10.0.0.4, 3345

Destination IP: 128.119.68.91, 80

Source IP: 10.0.0.4, 3345

Destination IP: 131.204.172.71, 5001

Source IP: 10.0.0.27, 3345

Destination IP: 128.119.68.91, 80

Question 41 1 / 1 pts

At Step 2, what are the source IP address, destination IP address and port numbers?

Source IP: 131.204.172.71, 5001

Destination IP: 10.0.0.4, 3345

Source IP: 10.0.0.27, 80

Destination IP: 131.204.172.71, 5001

Source IP: 10.0.0.27, 3345

Destination IP: 128.119.68.91, 80

Source IP: 131.204.172.71, 5001

Destination IP: 128.119.68.91, 80

Question 42 1 / 1 pts

At Step 3, what are the source IP address, destination IP address and port numbers?

Source IP: 128.119.68.91, 5001

Destination IP: 131.204.172.71, 3345

Source IP: 128.119.68.91, 80

Destination IP: 10.0.0.27, 3345

Source IP: 128.119.68.91, 80

Destination IP: 131.204.172.71, 5001

Source IP: 131.204.172.71, 5001

Destination IP: 10.0.0.4, 3345

Question 43 1 / 1 pts

At Step 4, what are the source IP address, destination IP address and port numbers?

Source IP: 128.119.68.91, 80

Destination IP: 10.0.0.4, 3345

Source IP: 10.0.0.27, 3345

Destination IP: 10.0.0.4, 3345

Source IP: 128.119.68.91, 80

Destination IP: 10.0.0.27, 3345

Source IP: 131.204.172.71, 5001

Destination IP: 10.0.0.4, 3345

Answer Question 44 using the following information.

Consider the scenario as shown in Figure 9 below.

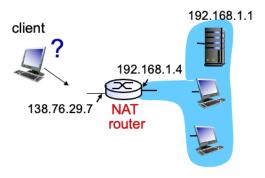


Figure 9

The client is in the outside world, and it is trying to connect to a server that is inside a local network, where a NAT router translates an address known to the outside world into the address of a host in the local network of an institution.

Question 44 1 / 1 pts

Which of the listed method can solve this problem?

Universal Plug and Play (UPnP)

Client creates connection to relay.

All of these listed

Statically configure NAT to forward incoming connection.

Answer Questions 45 to 47 using the following information.

Consider distance-vector algorithm.

Given the network shown in Figure 10 below and assume that each node initially knows the cost to each of its neighbors. Fill in the **FINAL** distance table entries at node z.

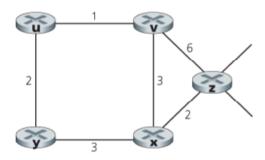


Figure 10.

			Cost to			
		u	v	x	у	z
From	v	1	0	3	3	5
FIOIII	x	4	3	0	3	2
	z	А	В	2	D	0

Table 1

Question 45	1 / 1 pts
What is the cost from z to u?	
○ ∞	
6	
None of the cost listed	
O 7	

Question 46	1 / 1 pts
What is the cost from z to v?	
None of the cost listed	
O 6	
5	
○ ∞	

Answer Questions 48 to 50 using the following information.

Consider the scenario shown in Figure 11 below. The number on the link represents the cost from one node to another.

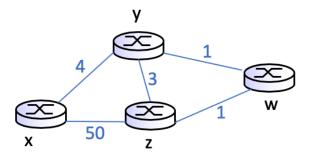


Figure 11.

Question 48 3 / 3 pts

Suppose after the distance vector routing is stabilized, the link cost between x and y increases to 60. What problem may occur?	
O F	Flooding problem
• (Count to infinity
	No route to host
O 1	Hot potato routing

Question 49	3 / 3 pts
Which method will sometimes solve the problem in Question 48	3?
Reverse path forwarding	
Cut link between y and z	
Broadcast cost increase	
Split horizon with poisoned reverse	

Under what condition will the method in Question 49 above Not solve the problem in Question 48? When there is a loop in the network graph

When a minimum spanning tree cannot be created
When multiple paths can be used
When there is a bottleneck node

Quiz Score: 92 out of 100