CSE 123: Computer Networks Fall Quarter, 2010 FINAL EXAM

Instructor: Alex C. Snoeren

Name	SOLUTIONS	
Student ID		

Question	Score	Points
1	20	20
2	15	15
3	15	15
4	30	30
5	30	30
6	30	30
7	10	10
Total	150	150

This exam is **closed book**. You are allowed one 8.5x11-inch, double-sided sheet of paper containing whatever you would like (a "crib sheet"). The exam contains seven questions of differing point values. Each question is clearly labeled with its value. Please answer all questions in the space provided. You have three hours to complete this exam. As with any exam, I suggest you read through all the questions first before answering any of them. Note that the final question is a freebie; you will receive full credit regardless of your answer. I would, however, appreciate your feedback if you have time remaining after completing the remainder of the exam.

GOOD LUCK!

_	ots) True/False. Determine whether each of the following statements is true or false. No explanais necessary; partial credit will not be awarded.
a)	All nodes connected to the Internet must implement UDP.
	False
b)	Channel noise leads to signal attenuation.
	False; just a drop in the SNR ratio.
c)	Media Access Control is a function of the data-link layer.
	True
d)	Switches decrement the TTL field in the IP header.
	False; only routers do.
e)	RTS/CTS combats the Hidden Terminal problem.
	True
f)	Token Ring is an example of a contention-based MAC protocol.
	False
g)	A one-bit parity scheme has Hamming distance 1.
	False; it has a Hamming distance of 2.
h)	FEC can be more efficient than ARQ in a broadcast environment with many receivers.
	True
i)	IS-IS is a distance vector routing protocol.
	False; it's link state.
j)	BGP exchanges link weights.
	False; BGP exchanges path vectors.

a)	Why are digital encoding schemes with more frequent transitions preferred?
	Assists in clock recovery and avoids baseline drift.
b)	What is the minimum value for the timeout of a reliable transmission protocol? Why?
	One RTT; anything shorter wouldd timeout before the ACK had a chance to arrive.
c)	How does 802.11 implement virtual carrier sense?
	It uses both the NAV and, optionally, RTS/CTS.
d)	What is the difference between congestion control and flow control?
	Congestion control prevents overrunning buffers in the network, while flow control prevents overflowing the receiver's.
e)	What are the drawbacks of sentinel-based framing?
	Sentinel-based framing requires the sender to do some form of stuffing, and the receiver to reverse the process. This adds both complexity and overhead to the transmission process, as well as decreasing the efficiency of the link.

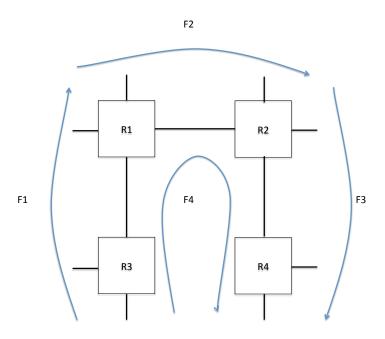
2. (15 pts) Short answer. Concisely answer each of the following questions.

a)	What is the difference between the bit rate and baud rate of a signal?
	Baud rate is the speed at which symbols can be communicated across a channel; bit rate depends on the information rate of each symbol. I.e., if each symbol corresponds to more than one bit, bit rate will exceed baud rate.
b)	Why does distance-vector routing scale better than link-state routing? Be specific.
	Routers in distance vector protocols only receive updates from their immediate neighbors, while routers must manage LSPs from every router in the network in link-state protocols.
c)	Explain the difference between traffic shaping and traffic policing.
	Traffic shaping controls the rate of sending by buffering excess incoming traffic, while traffic policing drops any excess.
d)	Give one advantage and one disadvantage of window-based flow control vs. rate-based flow control.
	Window-based flow control is simple to implement and does not require fine-grained clocks, but leads to more bursty behavior than rate-based flow control.
e)	List two reasons why intra-domain routing protocols are not suitable for inter-domain routing.
	1) ASes do not wish to expose their internal topology to other ASes, and 2) as discussed in part b), the number of messages each router has to process scales with the number of routers in the network, not the number of neighbors, so link-state will not work well in very large networks.

3. (15 pts) More short answer. Provide brief answers for the questions below.

4. (30 pts) Fair Queuing.

Answer the following questions about the network of four routers below. Each link has capacity of 1 Mbps. You can assume there is no contention on the access links, or for router backplane resources; i.e., the only constraints are the link capacities between routers. There are four flows in the network, labeled $F1\ldots F4$, that traverse the routers indicated. F4 shares links with every other flow, traversing $R3 \to R1 \to R2 \to R4$.



a) Assume each router implements FIFO queuing. If each flow consists of an identical, 1-Mbps constant bit rate UDP flow with equal packet sizes, what will the resulting rate be for each flow? You can assume that FIFO drops packets with uniform probability.

F1 and F4 compete at R3 1:1, giving each 1/2 Mbps. F4 then competes at R1 with F2 at 1/2:1, giving F2 2/3 Mbps and F4 1/3 Mbps. Hence, F4 competes at R2 with F3 at 1/3:1, resulting in a final allocation for F3 of 3/4 Mbps and 1/4 Mbps for F4.

b) Now consider the case where all routers implement fair queuing. What would be the throughput of each flow now?

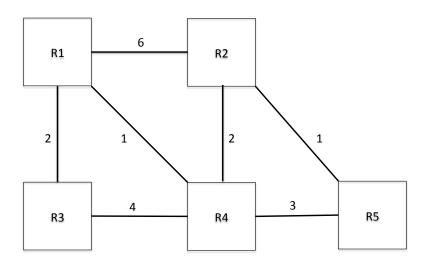
Every flow's fair share at each link is 1/2 Mbps. F4 will always be at least 1/2 Mbps when it competes with each other flow, so all flows get precisely their fair share: 1/2 Mbps each.

c) Finally, assume all routers implement WFQ, and each flow is assigned a weight equal to its number (i.e., flow F4 gets weight 4). What are the resulting throughputs?

F1 and F4 compete 1:4 at R3, giving F1 1/5 Mbps and F4 4/5 Mbps. F4 then competes with F2 at 4:2, so F2's fair share is 1/3 Mbps, giving F4 2/3 Mbps. Finally, F4 competes with F3 at 4:3, so F3's fair share is 3/7 Mbps, resulting in a final allocation for F4 of 4/7 Mbps.

5. (30 pts) Routing & Token buckets.

a) Consider the network of five routers below, with edge weights labeled.



Show the step-by-step operation of Dijkstra's algorithm for R3 using the table below.

Step	Confirmed	Tentative	Comments
1	(R3, 0, -)		
2	(R3, 0, -)	(R1, 2, R1) (R4, 4, R4)	
	(D2 0) (D1 2 D1)	(D) (D)	
3	(R3, 0, -)(R1, 2, R1)	(R4, 4, R4)	
4	(R3, 0, -)(R1, 2, R1)	(R4, 3, R1) (R2, 8, R1)	It is cheaper to reach $R4$ through $R1$
5	(R3, 0, -)(R1, 2, R1)	(R2, 8, R1)	
	(R4, 3, R1)		
6	(R3, 0, -)(R1, 2, R1)	(R2, 5, R1) (R5, 6, R1)	It is cheaper to reach $R2$ via $R4$
	(R4, 3, R1)		
7	(R3, 0, -)(R1, 2, R1)	(R5, 6, R1)	
	(R4, 3, R1) (R2, 5, R1)		
8	(R3, 0, -)(R1, 2, R1)	(R5, 6, R1)	R5 is the same cost through $R2$ or $R4$
	(R4, 3, R1) (R2, 5, R1)		
9	(R3, 0, -)(R1, 2, R1)		Done.
	(R4, 3, R1) (R2, 5, R1)		
	(R5, 6, R1)		

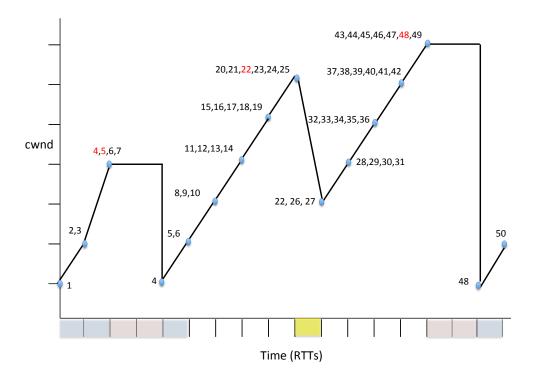
b) Now consider a Bellman-Ford-based distance-vector protocol that uses split horizon and poison reverse. Assume that the protocol has converged, and then node R5 fails. Will the protocol reconverge? If so, show the sequence of updates that result. (After each update, show the nexthop and cost for R5 as seen by each of the remaining routers.) If not, explain why not.

This will not converge, because the topology contains a cycle of three routers (e.g., R1, R3, R4) where only one is directly connected to the failed node, R5. (In fact, when analyzing this problem, you can safely remove R2 from the network as no router will ever use it as a nexthop for any other.) The remaining subset of the network looks just like the case on Slide 17 in lecture 13. Hence, the routers will count to infinity because they each only poison one of the others at a time.

c) Consider a token bucket with maximum rate R=20 Mbps. Suppose we want to make sure the maximum rate can only be sent for at most 5 seconds at a time, and at most 150 Mb can be sent over any 10-second window. Compute the required values for the token refresh rate, r, and the bucket depth, b.

If we send the maximum rate for 5 seconds, we send 20*5=100 Mb. We can only send 150 Mb in any 10-second window, including those we send at max rate, so 150-100=50 Mb, which we can send in the remaining 5 seconds, so r=50/5=10 Mbps. Now, in order to ensure the bucket has enough tokens to sustain a 5-second burst at 20 Mbps, we require b=(R-r)*5=(20-10)*5=50 Mb.

- 6. (30 pts) TCP congestion control.
 - a) (15 pts) Consider a TCP Reno (i.e., one that implements fast retransmit and recovery) flow that has exactly 50 segments to send. Assume that during the transmission, exactly four packets are lost: the 4th, 5th, 22nd, and 48th; no other losses occur. Using the graph below, plot the evolution of the congestion window as each segment is sent. You may measure cwnd and time in whatever units you find convenient. (Do not inflate the window due to duplicate ACKs.)



- **b)** (10 pts) Label your plot above indicating the regions where slowstart, timeout, congestion avoidance, and fast retransmit occur.
 - Timeouts, shown in red, occur after the loss of segments 4, 5, and 48. Slowstart periods, depicted in blue above, are until the first loss, and one RTT after each timeout. A fast retransmit of segment 22 is highlighted in yellow. Everything else is congestion avoidance. (Note I picked 2 RTTs for timeout value, you could use anything 1 RTT or greater.)
- **c)** (5 pts) Give one reason why TCP performs poorly when traversing overflowing FIFO queues, and explain how RED attempts to address the issue.
 - TCP flows are bursty, so when they encounter a full FIFO buffer, they will experience large numbers of losses in a window. RED spreads losses out. Also, TCP doesn't realize there's congestion until the queue is full, but RED will signal congestion earlier by dropping before the queue fills.

7.	(10 pts) You will receive full credit for this question regardless of how you respond, so DO NOT ANSWER IT UNTIL YOU ARE FINISHED with the remainder of the exam. There is no penalty if you don't get to it.
	a) What topic covered in this course did you find the most interesting?
	b) What topic did you find least interesting, and why?
	c) How long did you spend on each project? Which one did you prefer?
	d) Was there a networking topic you wish we had covered?
	e) Is there anything you'd suggest the professor to do differently next time he teaches this course?

Quizlet

Network design study guide

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92 terms











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Analyze requirements	Phase of network design does the network analyst	\Rightarrow
	interview users and technical personnel to gain an	
	understanding of the business and technical goals.	
Logical design	Phase of network design deals with network layer	☆
	addressing and switching and routing protocols	
Physical design	Phase of network design involves investigation of service providers.	☆
Phase involves building a prototype	test, optimize, and document the design	☆
Cladding	Reflective material surrounding the glass or plastic core of a fiber optic cable.	☆
Vertical	Wiring that runs between floors in a multi-floor structure.	☆
Horizontal	Wiring runs from a telecommunications closet to wall plates in offices.	☆
Segment	Raw data that is divided into a smaller unit; example 3877	☆



ROMMON	Low-level operating system commonly used for testing and troubleshooting on a Cisco router. CLI prompt in ROM Monitor mode.	
ifconfig	UNIX command will display the IP address and other configuration information about host's connection to the network.	☆
nslookup	Windows utility allows a user to query the configured name server to resolve a given host name. Network utility program used to obtain information about internet servers.	☆
ipconfig /displaydns	Windows command that will display the contents of the locally cached DNS entries. It is generally not necessary to view the contents of the DNS resolver cache, but this activity may be performed as a name resolution troubleshooting	\$
Fully qualified domain name (FQDN)	Attaching the name of a host to a domain name with a period in between them. host.domain example mystuff.somehighschool.edu	₩
Secure, monitor, test, improve	Network Security Wheel used for security policy compliance.	☆
File compression	Reduces the size of a file.	☆
Base lining	Create a starting point for ongoing evaluation of network health.	☆
Configuration Management	Records the operating system version which is installed on all switches and routers on your network.	☆
172.16.0.0 - 172.32.255.255	B class Private IP Address range	☆



Private IP address	A Class	
range	B Class	
	C Class	
	dress Range Default S	
) - 10.255.255.255	
	0.0 - 172.31.255.255 25 3.0.0 -	
	3.255.255 25	
Inside Local	NAT addressing referring to a host on the local network.	\Rightarrow
	Cisco defined.	
Inside global	A legitimate IP address assigned by the NIC or service	\Rightarrow
	provider that represents one or more inside local IP	
	addresses to the outside world. Cisco defined.	
Outside local	The IP address of an outside host as it appears to the	\Rightarrow
	inside network. Not necessarily a legitimate address, it is	
	allocated from an address space routable on the inside.	
	Cisco defined.	
Outside global	The IP address assigned to a host on the outside	\Rightarrow
	network by the host owner. The address is allocated	
	from a globally routable address or network space.	
	Cisco defined.	
Permissive security	Permits access through all ports except those explicitly	\Rightarrow
policy	denied.	
Gnutella	Protocol allows P2P (Peer to peer) applications to search	☆
	for shared resources on peers across the Interenet.	
Dark fiber	Fiber Optic cable that is not in use or has not been	\Rightarrow
	terminated.	
255.255.0.0	Subnet mask used to identify the range of class B private	\Rightarrow
	IP addresses.	
Backup process	Copying data to media that is not stored on the device.	☆
		\Rightarrow

Differential backup	Backup process archives all selected files that have changed since the last full or incremental backup and does not reset the archive bit.	
Full backup	A backup in which all of a defined set of data objects are copied, regardless of whether they have been modified since the last backup. Also resets the archive bit.	☆
Incremental backup	A security copy that contains only those files that have been altered since the last full backup.	☆
Network Attached storage	A specialized storage device of group storage devices that provides a centralized fault-tolerant data storage for a network.	☆
Raid	Redundant Array of Independent Disks, a disk subsystem that increases performance or provides fault tolerance or both.	☆
Raid level 0	Disk striping across independent hard disk drives. Increases speed performance but no fault tolerance. Lose one drive, lose all information.	☆
Raid 1	Disk mirroring and/or duplexing with the same information on two hard drives. Increases security but not speed.	☆
Raid 3	Data storage redundancy through a process involving writing data across multiple disks and using an ECC error correction code (parity) stored on a single disk.	☆
Raid 5	Block striping with distributed parity. Distributes parity across all drives rather than one drive. Provides fault tolerance and increase in read-write data transfer.	☆
DHCP request is broadcast.	To let all DHCP servers know which IP address is being accepted by the client.	☆



DHCP steps to addressing	1. Request or discover 2. returns unassigned IP address or offer 3. accepts the IP address by sending out an UDP broadcast to all network nodes 4. Server acknowledges the assigned IP address	
OSI model layers	Arranged from the lowest to the highest levels Physical, Data link, Network, Transport, Session, Presentation, Application.	☆
Physical Layer	Lowest layer consists of cable and connectors for constructing the network. Concerned only with moving the information through wires, fiber optic, and wireless	☆
Data Link Layer	How the raw data is packaged for transfer from one network interface card to another. Includes address of the source and destination and size of the file.	☆
Network Layer	Responsible for routing packets from one network to another using IP addressing format. Navigating between networks.	☆
Transport Layer	Layer of the OSI model has the role of providing error checking by sequencing packets and reassembling them into their correct order. Makes sure the packets are in the correct order or packet quality.	☆
Session Layer	Establishes connection between two different computers. and provides security based on computer and user name recognition. Like a hand shake at the beginning and end of the meeting.	☆
Presentation Layer	This ensurers character code recognition. Converting code to make it recognizable to each computer.	☆
Application Layer	Works with specific networking applications such as web browsers, file transfer programs, and e-mail.	☆
Network Layer	Routes packets from one network to another, navigating networks following a path of least traffic.	☆

Application Layer	Works with specific networking applications such as Web	\Diamond
	browsers, file transfer programs, and e-amil. Establishes communication with a network.	
Presentation Layer	Ensures character-code recognition. Responsible for converting character codes into a code that is recognizable by a computer that uses a different character codes. Example PC to mainframe changes ASCII to EBCDIC.	☆
Session layer	Establishes a connection between two different computers and provides security based on computer and user name recognition. Session layer and transport layer are sometimes combined.	☆
CSMA/CD	Type of media access method used by NICs that listen to or "sense" the cable to check for traffic and send only when they hear that no one else is transmitting. Computers take turns transmitting.	☆
"50 service" on an HP printer	Fuser on the HP printer causes this error	☆
Fuser in a printer	A pair of heated rollers within the printer that fuses the toner onto the paper being printed on. Toner particles are pulled to the paper to create the desired text and images as the paper passes the imaging drum. This will smudge until the toner is fused to the paper.	☆
Half-duplex	Limits network equipment, so it cannot send and receive at the same time.	☆
Full-duplex	Communication between two components means that both can transmit and receive information between each other simultaneously. example Telephones	☆
RJ-45	Connector used on a UTP network cable.	☆
		\Rightarrow

UTP	Unshielded Twisted Pair, cables made with twisted pair wire.	
EAP	Extensible Authentication Protocol, or EAP, is an authentication framework frequently used in wireless networks and point-to-point connections. Supports smart cards.	☆
Network administrators limit the size of failure domains because	Reduces the impact of a key device or service failure.	☆
BIND	(Berkely Internet Name Domain) is a popular software for translating domain names into IP addresses and usually found on Linux severs.	☆
DNS	Domain Name System, the Internet's system for converting alphabetic names into numeric IP addresses.	☆
Spanning Tree Protocol	A Layer 2 (Data Link) protocol that runs on bridges and switches. The specification for STP is IEEE 802.1D. The main purpose of STP is to ensure that you do not create loops when you have redundant paths in your network.	☆
IRQ	Interrupt request (PC architecture) is a hardware signal sent to the processor that temporarily stops a running program and allows a special program, an interrupt handler, to run instead. Such as a Network Interface Card NIC	☆
Configuration software for a NIC	Sends an IRQ to the processor to allow legacy ISA NIC to be used on a PC computer.	☆
Legacy card or network	Old hardware or software that is no longer supported on current protocols. Floppy disk are a legacy device. Netscape is a legacy browser	☆
GPO	Group Policy Object Windows organizes users into Groups to assign permissions	☆

Groups	Windows organizes user accounts into that	☆
	provide the same permissions to all members.	
The computer is not configured to use DNS	What does it mean if a network administrator can PING a resource server using its IP address, but not its Internet name?	☆
Fiber Optic cabling	Transfers information using light. Immune to electromagnetic or radio-frequency interference.	☆
ACL Statement	Access Control List - a network filter utilized by routers and some switches to permit and restrict data flows into and out of network interfaces.	☆
Implicit deny any statement placed at the end of the list will cause the packets to be dropped.	ACL statements operated in sequential, logical order. If a condition match is true, the rest of the statements are not checked. If all of the ACL statements are unmatched, what happens to the packet?	☆
Layer 2 of an OSI model	Workgroup network switches typically function at this layer of the OSI model.	\Rightarrow
10000000.0000001.1000 0000.0000111	Binary representation of 128.1.128.7	☆
Tracert	TCP/IP utility used to check the pathways and status of network devices between two Internet paths/sites.	☆
Hardware RAID	Does not require the operating system to handle most of the data striping and parity calculations.	☆
Zone	Create by configuring DNS with a new Domain entry.	\Rightarrow
Last Sunday and Tuesday	The administrator has set a backup routine that includes differential backups on Monday through Friday with a full (normal) backup on Sunday. Your system crashes on Wednesday morning. What tapes will be needed for the restore.	☆

Wrong printer driver	Your client reports that when they print their word	\Rightarrow
	processing documents it appears on the printer as X!####\$DF()+~~ and continues to print numerous pages with random characters. What is the most likely problem.	
Printer driver	Small computer program that enables an application software (word processor, spreadsheet, graphics program, etc.) to communicate with a printer. Every printer brand (and every model under that brand) has its unique driver software for a specific operating system which must be installed for the computer to use the printer. Read more: http://www.businessdictionary.com/definition/printer-driver.html	☆
FTP	File Transfer Protocol is a standard Internet protocol for transmitting files between computers on the Internet over TCP/IP connections.	\$
IP	Internet protocol, the method by which information is sent between any two Internet computers on the Internet.	☆
SMTP	Simple Mail Transfer Protocol, a protocol for sending e- mail messages between servers.	☆
POP	POP, Post Office Protocol, is an internet standard that defines an email server (the POP server) and a way to retrieve mail from it (using a POP client).	\Rightarrow
TFTP	Trivial File Transfer Protocol. It's a technology for transferring files between network devices, and is a simplified version of FTP (File Transfer Protocol).	☆



TCP	Transmission Control Protocol is a standard that defines how to establish and maintain a network conversation via which application programs can exchange data. TCP works with the Internet Protocol (IP), which defines how computers send packets of data to each other. Together, TCP and IP are the basic rules defining the Internet.	
What does a switch do when it receives a frame and the source address is unknown?	Switch will map the source address to the port on which it was received.	\Rightarrow
Switch	A computer networking device that connects devices together on a computer network by using packet switching to receive, process, and forward data to the destination device.	☆
Which is the first step to test of destination reachability?	Echo request issued by source.	\Rightarrow
Bits per second	Units used to measure Internet bandwidth.	☆
Ports 1025 and above	Port numbers used by PAT to create unique global addresses.	☆
PAT	Port Address Translation, is an extension to network address translation (NAT) that permits multiple devices on a local area network (LAN) to be mapped to a single public IP address. The goal of PAT is to conserve IP addresses.	☆



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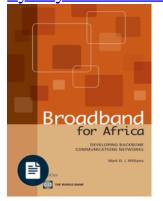
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Best Books About Transmission Control Protocol

• TCP/IP Sockets in Java: Practical Guide for Programmers

by Kenneth L. Calvert and Michael J. Donahoo

Practical TCP/IP and Ethernet Networking for Industry

by **Deon Reynders** and **Edwin Wright**

Essential Skills for Hackers

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Exam Source
<u>Question 1</u> : (QID=2961)
We want to assign port 3 of our Catalyst 2950 switch to VLAN 3. What command will achieve this?
1. switch(config)#switchport access vlan 3
2. switch#switchport access vlan 3
3. switch(config-if)#switchport vlan 3
4. switch(config-if)#switchport access vlan 3
Correct answer(s): 4
Explanation: The switchport access vlan [number] command will configure a switch port to that VLAN. You can also use the switchport access vlan dynamic command to configure the switch port automatically in a certain VLAN depending on the VLAN membership of the incoming packets. For both commands to work properly the switch port needs to be in access mode.
[Implement a small switched network]
Question 2: (QID=25148)
Which command will allow only ssh connections to the router?
1. transport input telnet ssh
2. transport input ssh
3. transport ssh
4. transport ssh enable
Correct answer(s): 2
Explanation: The transport input ssh line configuration commands allows ssh connections only to the router.

[Implement a small switched network]

<u>Question 3</u> : (QID=2986)
Which utility would we use to test all 7 OSI layers?
ı. Ping
2. Trace route
3. Telnet
4. Show arp
Correct answer(s): 3
Explanation: If we can telnet from one device to the other we have proven that the routing (layer 3) and the physical layer are working properly. Since telnet is an application layer utility it gives us confirmation that all 7 OSI layers are working correctly between the two devices.
[Implement a small switched network]
Question 4: (QID=1286)
You are the network administrator and are given the tasks to plan for the future expansion of the network. You decide to subnet the current network of 172.15.0.0. What will be the subnet mask be if every subnet has to allocate 500 hosts.
1. 255.255.0.0
2. 255.0.0.0
3. 255.255.240.0
4. 255.255.254.0
Correct answer(s): 4
Explanation: The requirement of having to allocate 500 hosts on each subnet can be achieved by the following: $2^9 = 512$. So deducting the broadcast and network address we have 510 possible hosts on each subnet. We used 9 bits to allocate the hosts, this leaves us 7 bits for possible subnets. Those 7 bits in binary is 1111 1110 or 254 in decimal.
Implement an IP addressing scheme and IP services to meet network requirements for a small branch office]
<u>Question 5</u> : (QID=2359)
What address range is commonly used for testing purposes?

2. Any available IP Address
3. 127.0.0.0/8
4. 240.0.0.0 – 255.255.255
Correct answer(s): 3
Explanation: The 127.0.0.0/8 address range in the Class A range is a reserved address range and used for testing purposes.
[Implement an IP addressing scheme and IP services to meet network requirements for a small branch office]
<u>Question 6</u> : (QID=2975)
If we have multiple IOS versions on our router, how can we view which ones are available?
1. Show version
2. Show memory
3. Show file systems
4. Show flash:
Correct answer(s): 4
Explanation: The show flash: command displays the contents of the flash file system.
[Implement a small routed network]
<u>Question 7</u> : (QID=2375)
What do you use when you want to display a message to all the terminals connected to a router?
ı. banner login
2. banner motd
3. banner exec
4. banner slip-ppp
Correct answer(s): 2

1. 10.0.0.0/16

Explanation: This MOTD banner is displayed to all terminals connected and is useful for sending messages that affect all users (such as impending system shutdowns). Use the no exec-banner or no motd-banner command to disable the MOTD banner on a line. The no exec-banner command also disables the EXEC banner on the line. When a user connects to the router, the MOTD banner appears before the login prompt. After the user logs in to the router, the EXEC banner or incoming banner will be displayed, depending on the type of connection. For a reverse Telnet login, the incoming banner will be displayed. For all other connections, the router will display the EXEC banner. Connection, Menu, and System Banner Commands

[Implement a small switched network]
----Question 8: (QID=1294)

Given the following IP Address 172.16.0.0/27 what is the 4th subnet's network address, host range and broadcast address? The router is running IOS version 12.2.

- 1. Subnet address 172.16.1.0, host range 172.16.1.1 172.16.1.30, broadcast address 172.16.1.31
- 2. Subnet address 172.16.0.128, host range 172.16.0.129 172.16.0.158, broadcast address 172.16.0.159
- 3. Subnet address 172.16.0.96, host range 172.16.0.97 172.16.0.126, broadcast address 172.16.0.127
- 4. Subnet address 172.16.1.128, host range 172.16.1.129 172.16.1.158, broadcast address 172.16.0.159

Correct answer(s): 3

Explanation: The default subnet mask for a class B address is /16 so we will have 11 bits for subnets.

12.2 is the IOS mentioned and that has the IP Subnet-Zero command enabled by default.

So the total amount of subnets is $2^{11} = 2048$ subnets. On each subnet we will have $2^{5} - 2 = 30$ hosts on each subnet.

Now we will calculate the subnet ranges. The following table is the binary version of the last octet.

subnet bits host bits IP address role

000 00000 172.16.0.0 1st subnet network address

000 00001 172.16.0.1 host # 1

000 00010 172.16.0.2 host # 2

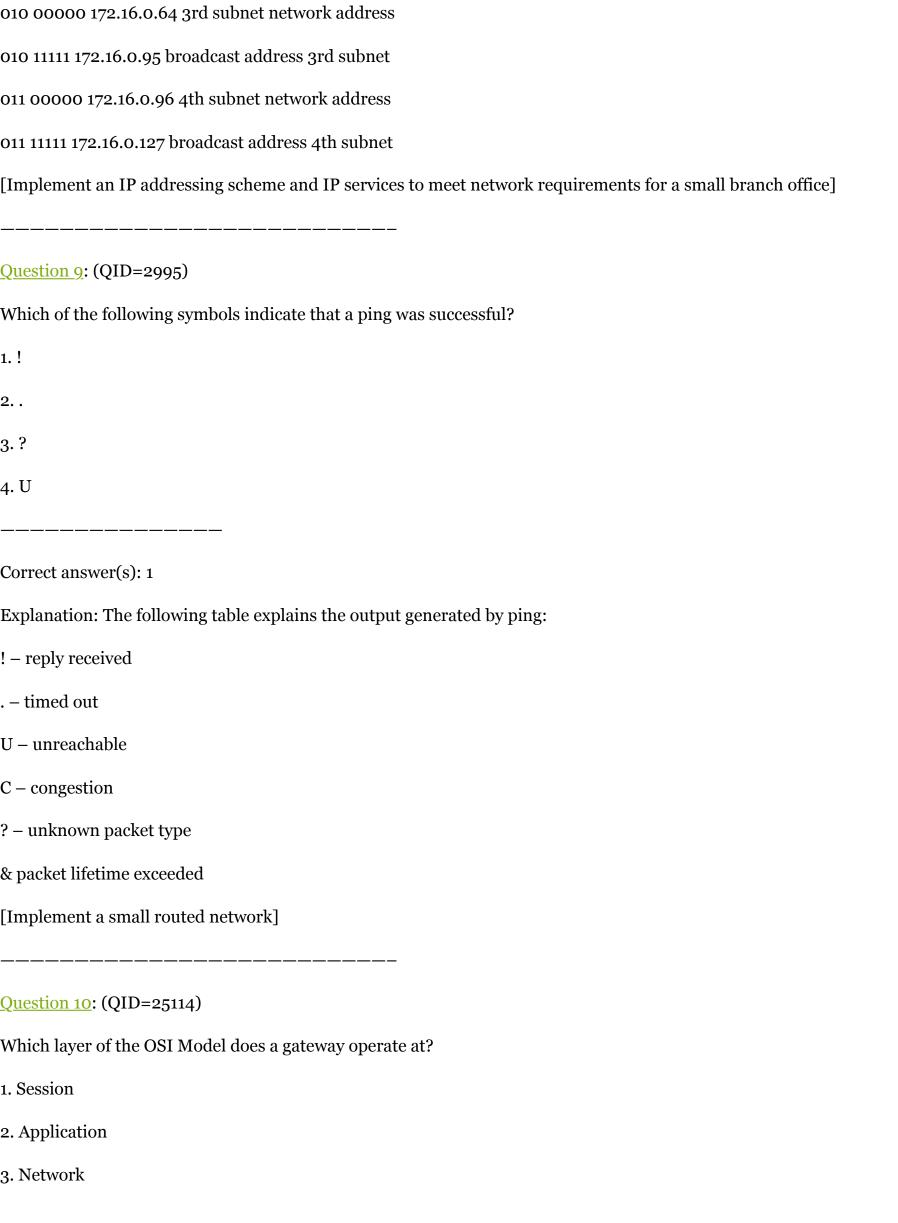
000 11110 172.16.0.30 host # 30

000 11111 172.16.0.31 broadcast address for 1st subnet

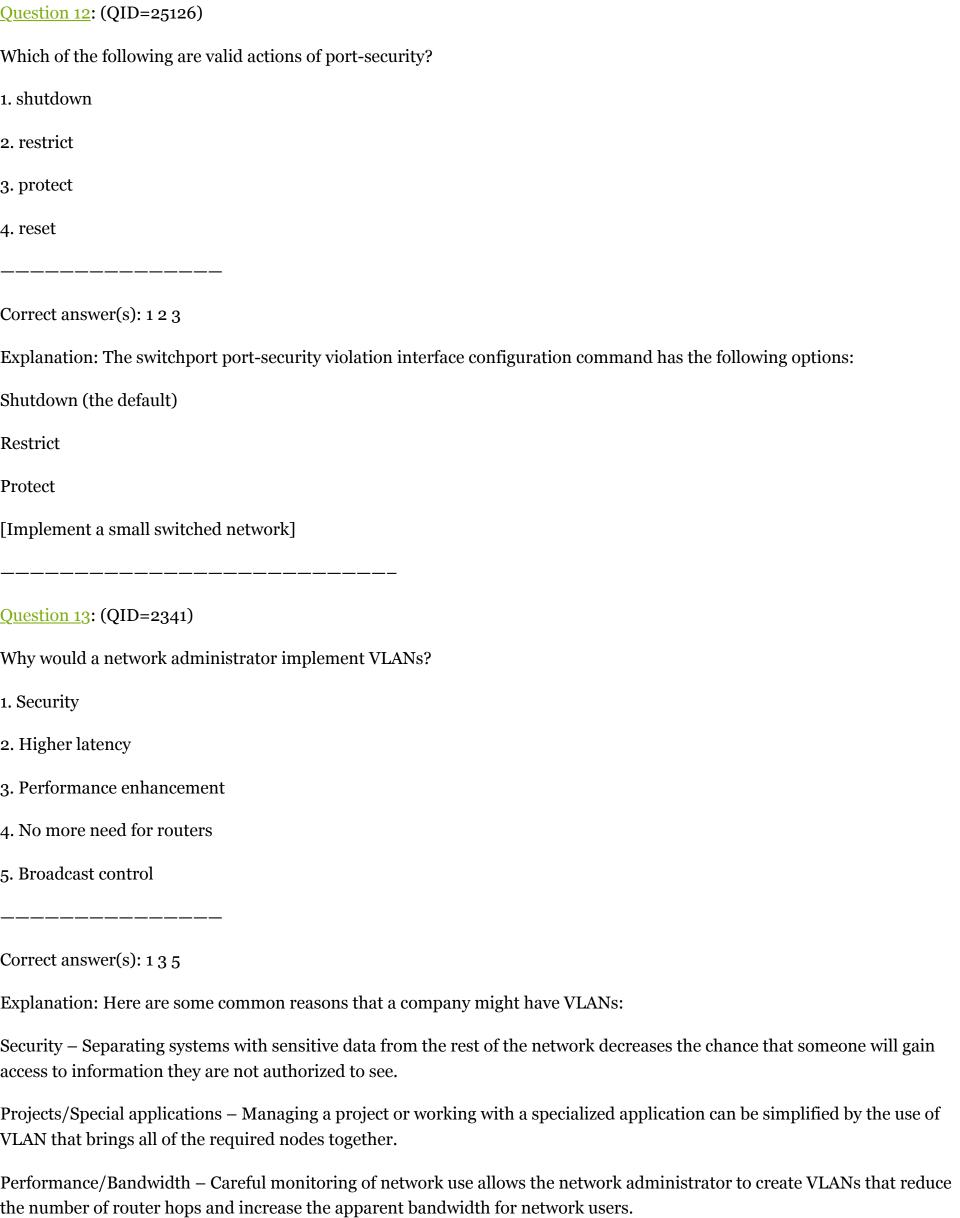
001 00000 172.16.0.32 2nd subnet network address

001 00010 172.16.0.34 host # 2 on 2nd subnet

001 11111 172.16.0.63 broadcast address for 2nd subnet

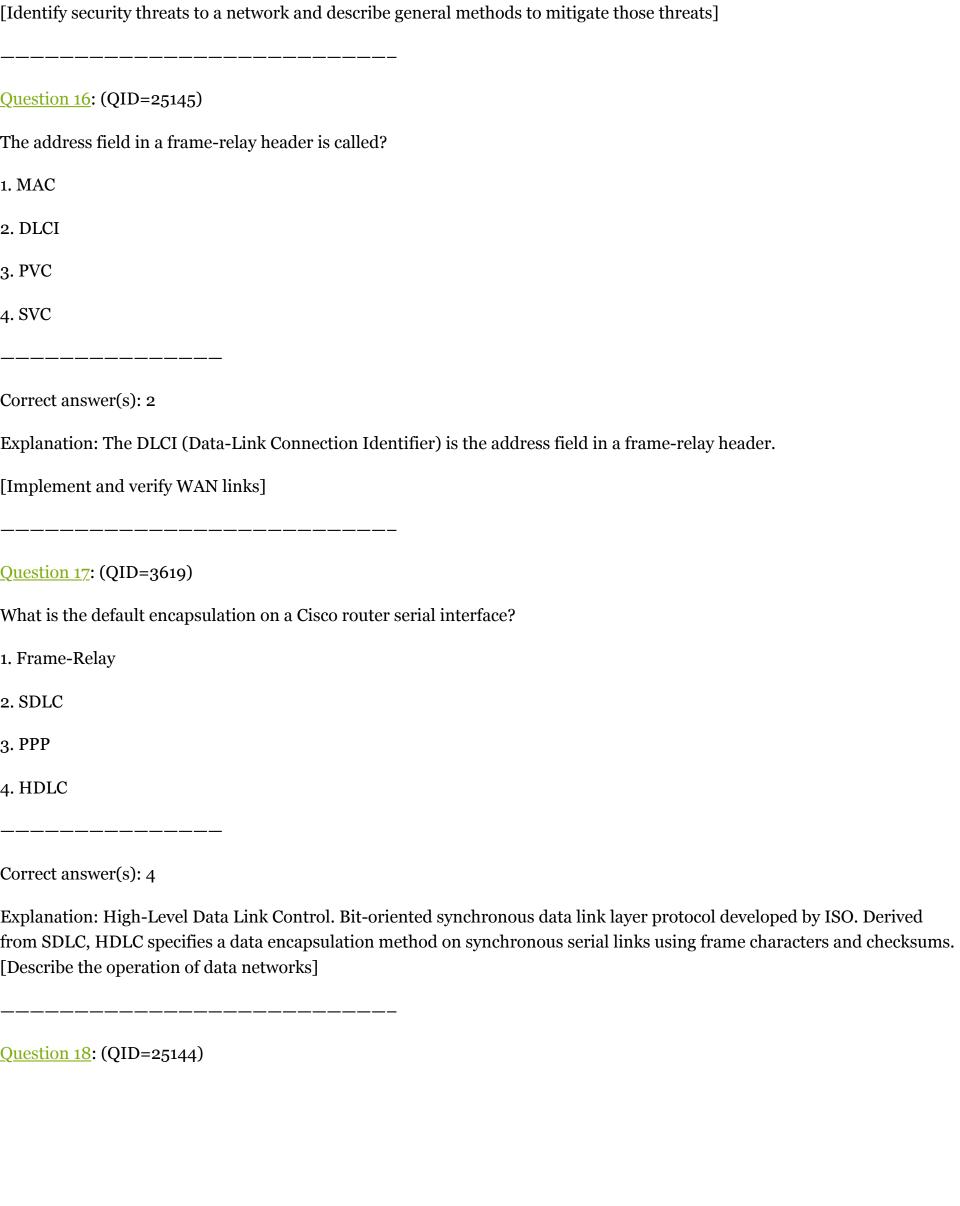


4. Datalink
5. Physical
Correct answer(s): 2
Explanation: [Describe the operation of data networks]
Question 11: (QID=3629)
What are the 2 sublayers of the Data Link layer of the OSI model?
1. Logical Layer Control
2. Logical Link Control
3. Media Access Control
4. Medium Access Control
Correct answer(s): 2 3
Explanation:
Layer Name Protocol/Services Details
7 Application Telnet, FTP, WWW, SMTP Provides and interface by which user applications access the network
6 Presentation ASCIIm, EBCDICm JPG, MIDI Encryption, compression and translation
5 Session RPC, NetBIOS Establishing, maintaining and mananging communication sessions
4 Transport TCP, UDP, SPX Flow and error control but for the end-to-end connection of 2 devices whose logical connection traverses a series of data links
3 Network IP, IPX, ARP, ICMP Routing of data across logical network paths is enabled here and defines packet and addressing format
2 Data Link (Sub-layers are Media Access Control and Logical Link Control) Frame Relay, LAPB, PPP, Ethernet, Token Ring, FDDI Controls how the medium is shared and accessed, how the devices are identified, how data is framed before transmission on the medium
1 Physical Electrical/optical protocols Voltage levels, timing of voltage changes, physical data rates, maximum transmission distances, physical connectors and other similar attributes are defined here
[Describe the operation of data networks]



are not part of the VLAN, it automatically reduces broadcasts. Access lists provide the network administrator with a way to control who sees what network traffic. An access list is a table the network administrator creates that lists what addresses have access to that network.
Departments/Specific job types – Companies may want VLANs set up for departments that are heavy network users (such as Multimedia or Engineering) or a VLAN across departments that is dedicated to specific types of employees (such as manager or sales people).
[Implement a small switched network]
Question 14: (QID=2354)
Which of the following describe full-duplex Ethernet?
1. it uses point-to-point connections
2. is limited to speeds of 100Mb
3. stations can transmit and receive at the same time
4. it uses point-to-multipoint connections
Correct answer(s): 1 3
Explanation: Using full-duplex Ethernet makes it possible for devices to transmit and receive at the same time which results in bandwidth optimization. Point-to-point connections are used between the device and switch. The IEEE802.x committee designed a standard for full duplex that covers 10BaseT, 100BaseX and 1000BaseX.
[Implement a small switched network]
Question 15: (QID=25140)
Which of the following would you use if you wanted to transfer data securely across the internet?
1. VPN
2. Leased Line
3. URL Filtering
4. Anti-x
Correct answer(s): 1
Explanation: A VPN is used to securely transfer data across insecure media, like the Internet.

Broadcasts/Traffic flow – Since a principle element of a VLAN is the fact that it does not pass broadcast traffic to nodes that





traverses a series of data links

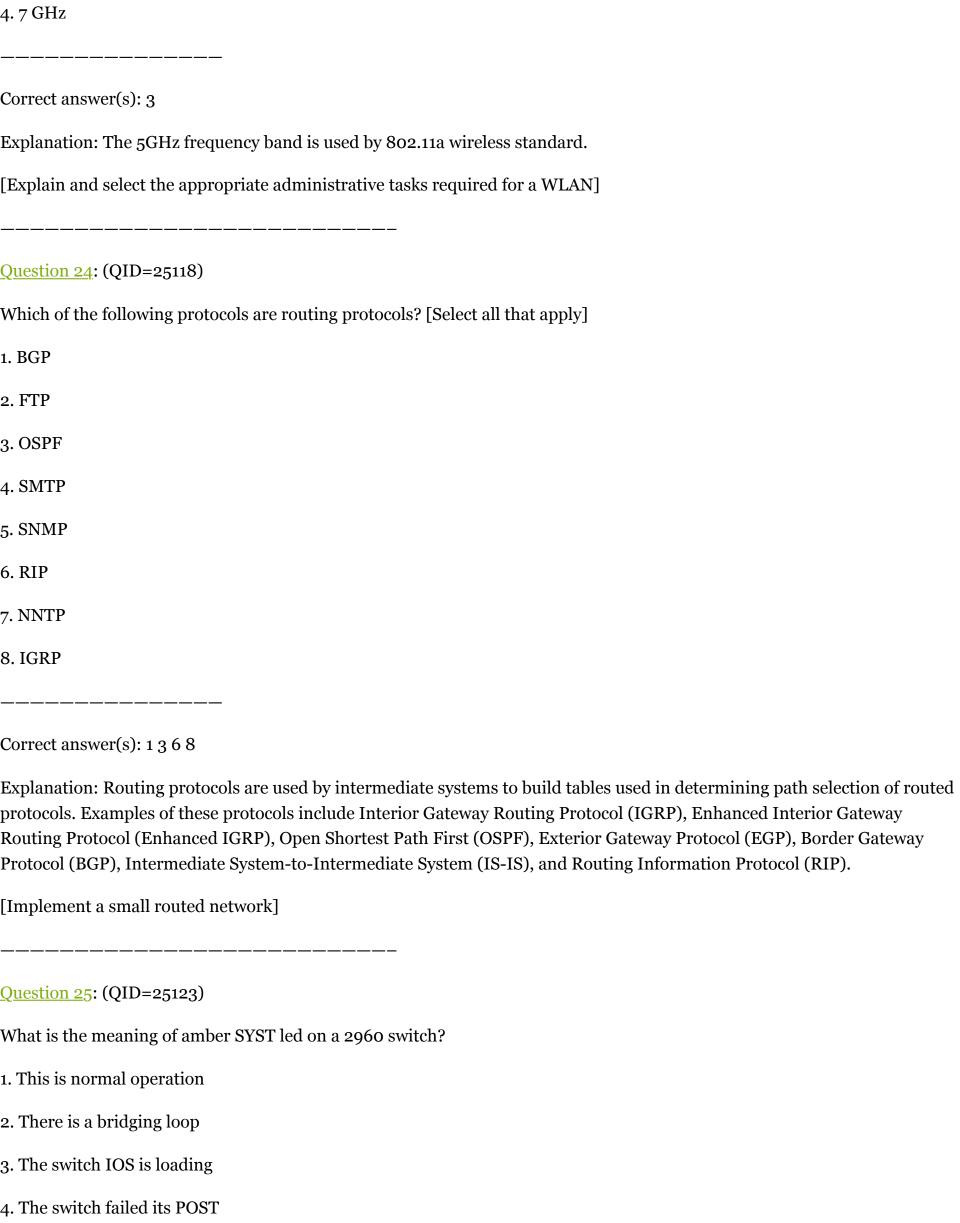
What type of cable is shown in the exhibit?
1. AEIA/TIA-232
2. EIA-530
3. X.21
4. V.35
Correct answer(s): 4
Explanation: [Implement and verify WAN links]
Question 19: (QID=1282)
Which of the following OSI layers is responsible for reliable transmission of data segments?
1. Session
2. Application
3. Transport
4. Data-Link
Correct answer(s): 3
Explanation: Layer Name Protocol/Services Details
7 Application Telnet, FTP, WWW, SMTP Provides and interface by which user applications access the network
6 Presentation ASCIIm, EBCDICm JPG, MIDI Encryption, compression and translation
5 Session RPC. NetBIOS Establishing, maintaining and mananging communication sessions

3 Network IP, IPX, ARP, ICMP Routing of data across logical network paths is enabled here and defines packet and

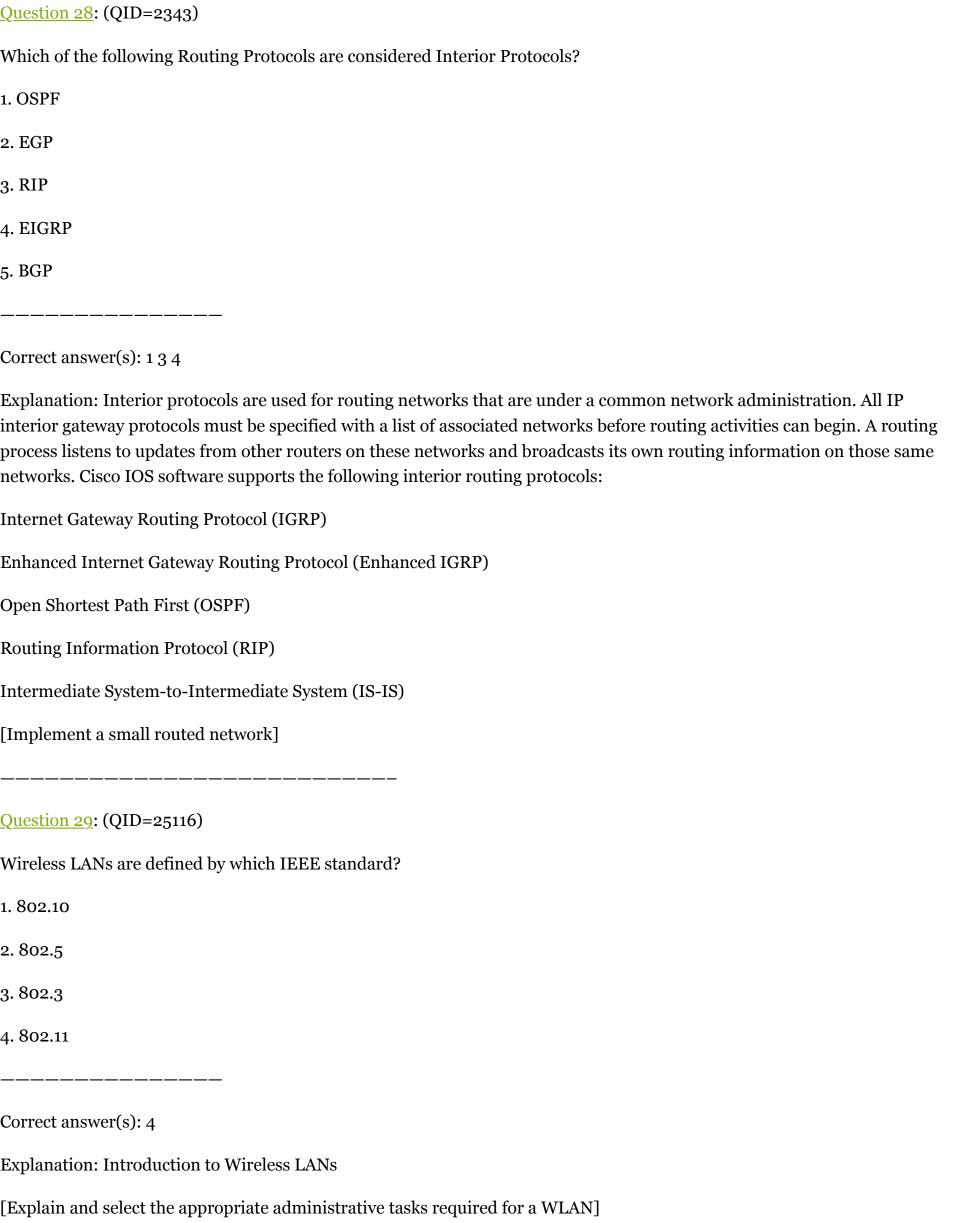
4 Transport TCP, UDP, SPX Flow and error control but for the end-to-end connection of 2 devices whose logical connection

addressing format
2 Data Link (Sub-layers are Media Access Control and Logical Link Control) Frame Relay, LAPB, PPP, Ethernet, Token Ring, FDDI Controls how the medium is shared and accessed, how the devices are identified, how data is framed before transmission on the medium
1 Physical Electrical/optical protocols Voltage levels, timing of voltage changes, physical data rates, maximum transmission distances, physical connectors and other similar attributes are defined here
[Describe the operation of data networks]
Question 20: (QID=1307)
Which of the following protocols operate at Layer 3 of the OSI model? [Select all that apply]
1. IP
2. TCP
3. ICMP
4. UDP
5. ARP
6. SPX
Correct answer(s): 1 3 5
Explanation: Layer Name Protocol/Services Details
7 Application Telnet, FTP, WWW, SMTP Provides and interface by which user applications access the network
6 Presentation ASCIIm, EBCDICm JPG, MIDI Encryption, compression and translation
5 Session RPC, NetBIOS Establishing, maintaining and mananging communication sessions
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1 Physical Electrical/optical protocols Voltage levels, timing of voltage changes, physical data rates, maximum transmission distances, physical connectors and other similar attributes are defined here
[Describe the operation of data networks]

<u>Question 21</u> : (QID=1297)
What command will change the hostname of a router?
1. router#host name new name
2. router(config-router)#hostname new name
3. router>host name new name
4. router(config)#hostname new name
Correct answer(s): 4
Explanation: To specify or modify the host name for the network server, use the hostname command in global configuration mode.
[Implement a small switched network]
Question 22: (QID=2362)
What command do we use to find out how long our router has been online?
1. Show version
2. Show hardware
3. Show uptime
4. Show history
Correct answer(s): 1
Explanation: The results of the show version command are shown below:
[Implement a small routed network]
<u>Question 23</u> : (QID=25128)
802.11a uses which frequency range?
1. 900 MHz
2. 2.4 GHz
3. 5 GHz

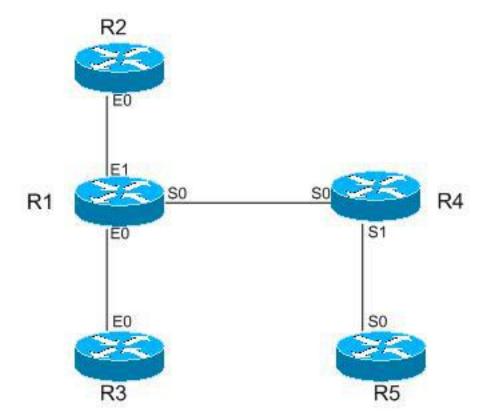


Correct answer(s): 4
Explanation: When the switch failed its POST, Power-On Self Test, the SYST led will turn amber and the IOS did not load.
[Implement a small switched network]
<u>Question 26</u> : (QID=2379)
We want to go back to user EXEC from privileged EXEC. What command do we use to achieve this without exiting the session?
ı. Exit
2. Log out
3. Disable
4. Logout
Correct answer(s): 3
Explanation: Click here for more information.
[Implement a small switched network]
<u>Question 27</u> : (QID=3632)
As a network administrator we want to make it easy to assign IP addresses to hosts. What protocol do we use?
ı. DNS
2. DHCP
3. HSRP
4. NTP
Correct answer(s): 2
Explanation: DHCP, as described by RFC 1541, or Dynamic Host Configuration Protocol is used for dynamically assigning network addresses and host specific configuration parameters.
[Implement an IP addressing scheme and IP services to meet network requirements for a small branch office]



<u>Question 30</u> : (QID=3685)
What are some advantages of having a layered OSI Model? [Select all that apply]
1. Changes at one layer do not affect the other layers
2. Networks are harder to design
3. Different vendor's equipment can easily interoperate
4. Changes at one layer are propagated throughout the other layers
5. Prevents routing loops
Correct answer(s): 1 3
Explanation: The main goal of the OSI model is to allow different vendors' networks to interoperate. Some benefits of the OSI model are:
changes at a certain layer do no affect the other layers.
various types of hardware & software can communicate with each other.
networks are more extensible.
[Describe the operation of data networks]
<u>Question 31</u> : (QID=3696)
Given the following network range 192.168.0.0/24. We want minimum 50 hosts on each of the available subnets. What will be the subnet mask?
1./24
2./25
3. /26
4. /27
Connect angreen(a): 0
Correct answer(s): 3
Explanation: 50 hosts require 6 bits of the last octet giving us 64 -2 = 62 possible hosts this satisfies our requirement and leaves us 2 bits over for the subnets. The default mask is /24 or 255.255.255.0 adding the 2 bits for the subnets it becomes 255.255.255.192 or /26.
[Implement an IP addressing scheme and IP services to meet network requirements for a small branch office]

Question 32: (QID=1299)
Which key-sequence would you use to move the cursor to the beginning of the line?
i. CTRL+ P
2. CTRL+ E
3. CTRL + A
4. CTRL + F
Correct answer(s): 3
Explanation: Keystrokes Purpose
Press Ctrl-B or press the left arrow key Moves the cursor back one character
Press Ctrl-F or press the right arrow key Moves the cursor forward one character
Press Ctrl-A Moves the cursor to the beginning of the command line
Press Ctrl-E Moves the cursor to the end of the command line
Press Esc B Moves the cursor back one word
Press Esc F Moves the cursor forward one word
Ctrl-P or the up arrow key Recalls commands in the history buffer, beginning with the most recent command. Repeat the key sequence to recall successively older commands
Ctrl-N or the down arrow key Returns to more recent commands in the history buffer after recalling commands with Ctrl-P or the up arrow key. Repeat the key sequence to recall successively more recent commands
[Implement a small switched network]
Question 33: (QID=1301)



You are the administrator of network X. You decided to implement RIP in the following topology (click exhibit). You do not want R3 to receive RIP updates. What configuration changes need to be implemented?

- 1. Implement a passive-interface on R1
- 2. Nothing needs to be configured
- 3. Create a sub-interface on R1
- 4. Change the routing protocol to OSPF

Correct answer(s): 1

Explanation: To block RIP broadcasts on an interface connected to a subnet of a RIP-enabled network add the passive-interface command to the RIP Process.

[Implement a small routed network]

<u>Question 34</u>: (QID=25124)

How do you enable SSH on your switch?

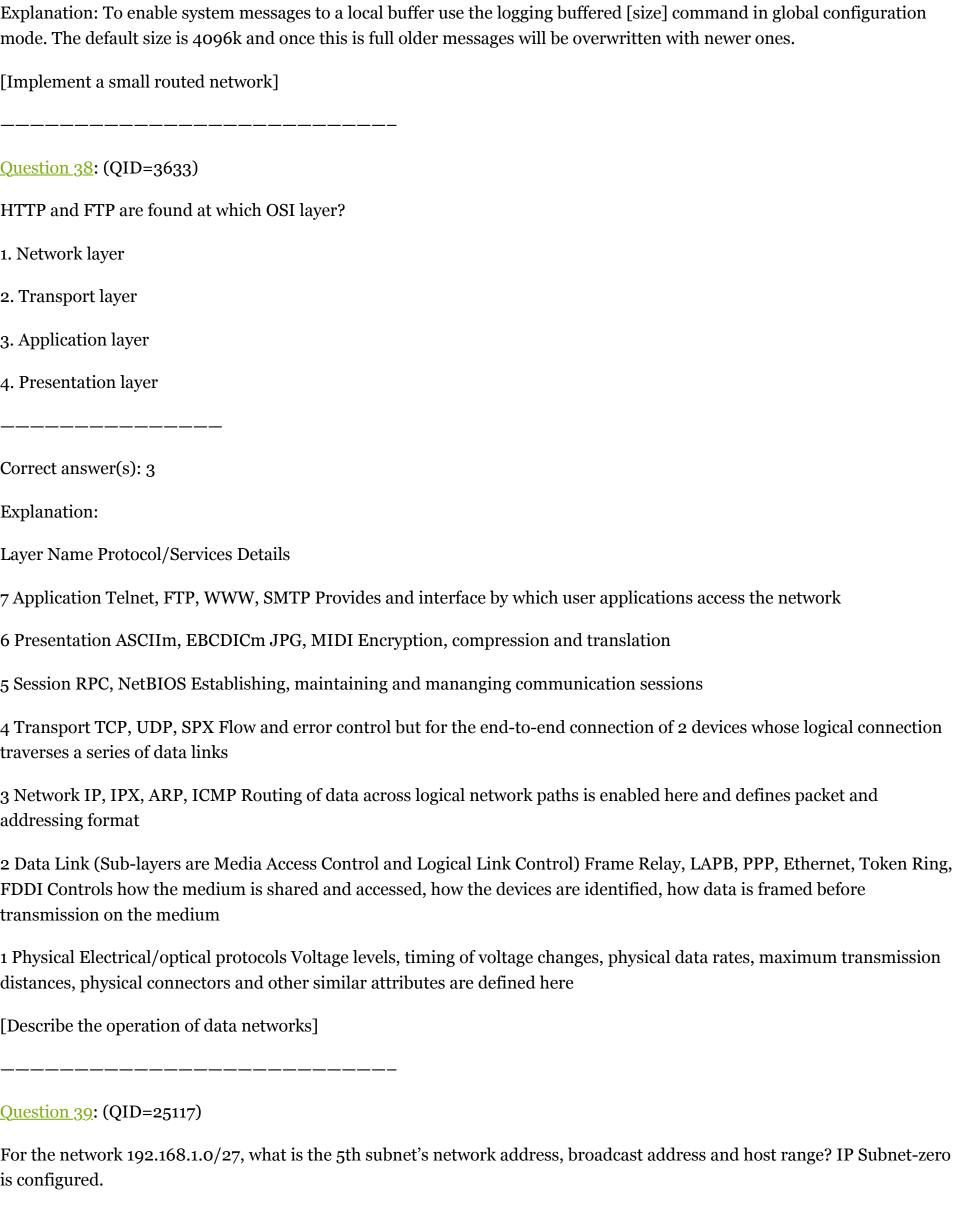
- 1. switch(config-line)#input ssh
- switch(config-line)#transport input ssh
- switch(config-line)#transport ssh
- 4. SSH is enabled by default

Correct answer(s): 3 Explanation: The transport input ssh command will allow SSH sessions to your switch. You also need to create a user and password and generate a public and private key pair. [Implement a small switched network] Question 35: (QID=25136) When we have a variation in delay it is called? 1. Loss Latency 3. Delay 4. Jitter Correct answer(s): 4 Explanation: Jitter occurs when there is a variation in the delay. [Describe the operation of data networks] <u>Question 36</u>: (QID=2998) E0 DTE DCE After connecting two routers back-to-back we are unable to ping across the serial connection. A closer look reveals that both serial interfaces have the following status: 'Serialo is up, line protocol is down' What needs to be configured on one of the routers to bring the serial interface up? [See the exhibit for network topology] A) router1(config)# interface serial o router1(config-if)#clock rate 64000

B) router1(config)#interface serial o

C) router2(config)#interface serial o
router2(config-if)#clockrate 64000
D) router2(config)#interface serial o
router2(config-if)#clock rate 64000
1. A
2. B
3. C
4. D
Correct answer(s): 4
Explanation: When connecting router back-to-back with a DTE/DCE cross-over cable we need to set the clocking on the DCE side of the connection. The show controllers serial [interface nr] command tells you what side is DTE or DCE.
[Implement and verify WAN links]
<u>Question 37</u> : (QID=3002)
Router#show logging Syslog logging: enabled (0 messages dropped, 1 messages rate-limited, 0 flushes, 0 overruns) Console logging: level debugging, 13 messages logged Monitor logging: level debugging, 0 messages logged Buffer logging: disabled Logging Exception size (4096 bytes) Count and timestamp logging messages: disabled Trap logging: level informational, 17 message lines logged Router#
We try to look at the log on the router but we get the following [see exhibit]. What do we need to do in order to view the log?
1. Reboot the router
2. Use the logging buffered global configuration command
3. Use the enable log all global configuration command
4. We can not do anything since there is no such thing as a log
Correct answer(s): 2

router1(config-if)#clockrate 64000

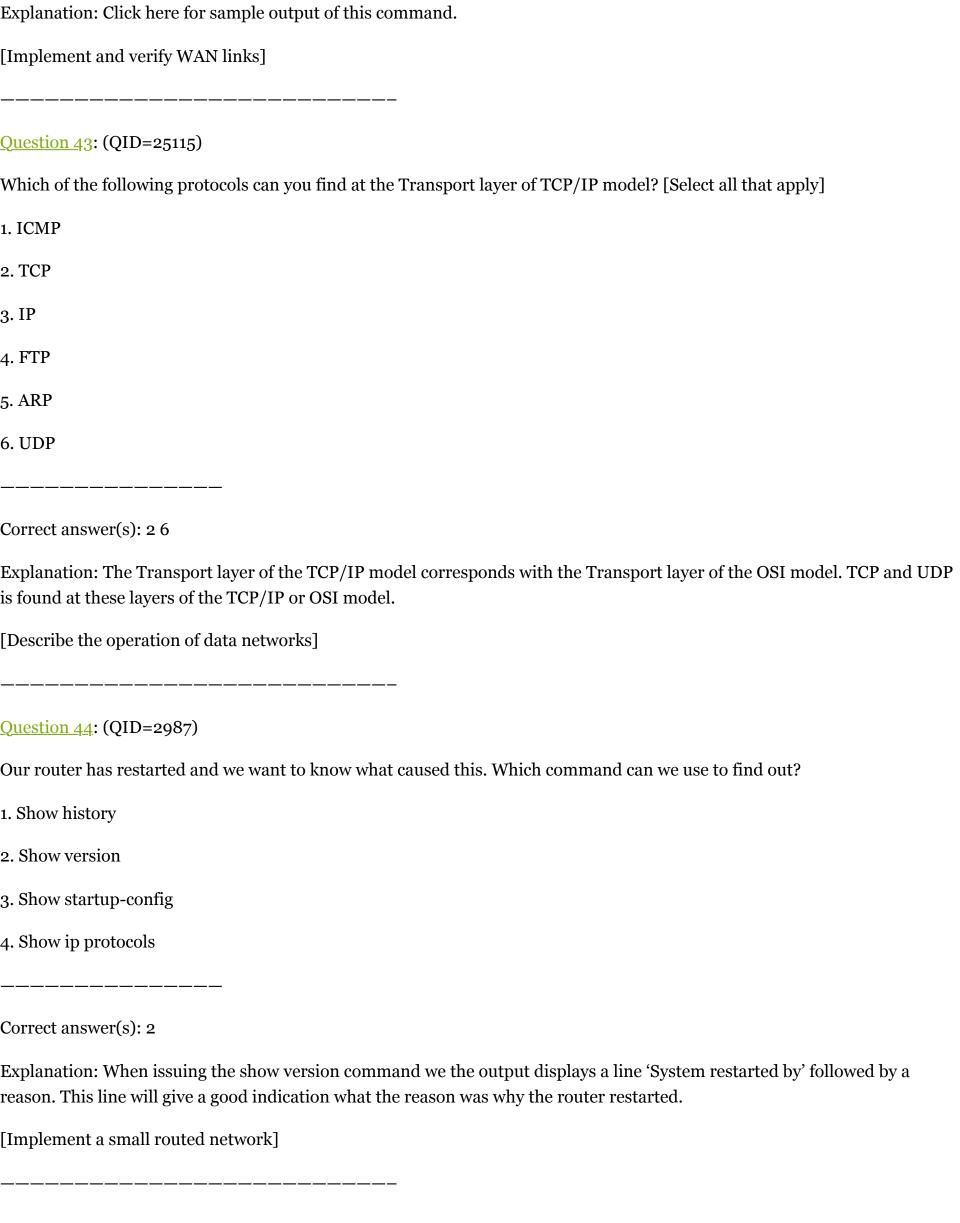


```
2. Subnet 192.168.1.160, Broadcast 192.168.1.191, hosts 192.168.1.161 – 192.168.1.190
3. Subnet 192.168.1.96, Broadcast 192.168.1.127, hosts 192.168.1.97 – 192.168.1.126
4. Subnet 192.168.1.128, Broadcast 192.168.1.159, hosts 192.168.1.129 - 192.168.1.158
Correct answer(s): 4
Explanation: The default subnet mask for a class C address is /24 so we will have 3 bits for subnets.
So the total amount of subnets is 2^3 = 8 subnets
On each subnet we will have 2^5 - 2 = 30 hosts on each subnet
Now we will calculate the subnet ranges:
The following table is the binary version of the last octet.
Subnet bits Host Bits IP Address Role
000 00000 192.168.1.0 1st subnet network address
000 00001 192.168.1.1 host nr 1
000 00010 192.168.1.2 host nr 2
000 11110 192.168.1.30 host nr 30
000 11111 192.168.1.31 broadcast address for 1st subnet
001 00000 192.168.1.32 2nd subnet network address
001 00010 192.168.1.34 host nr 2 on 2nd subnet
001 11111 192.168.1.63 broadcast address for 2nd subnet
011 00000 192.168.1.96 4th subnet network address
011 11111 192.168.1.127 broadcast address 4th subnet
100 00000 192.168.1.128 5th subnet network address
100 11111 192.168.1.159 broadcast address 5th subnet
[Implement an IP addressing scheme and IP services to meet network requirements for a small branch office]
<u>Question 40</u>: (QID=25137)
What kind of attack is designed to gather information?
```

1. Subnet 192.168.1.0, Broadcast 192.168.1.31, hosts 192.1683.1.1 – 192.168.1.30

2. DDoS attack
3. Reconnaissance attack
4. Access attack
Correct answer(s): 3
Explanation: A reconnaissance attack is an attack aimed at gathering information as preparation for further attacks.
[Identify security threats to a network and describe general methods to mitigate those threats]
<u>Question 41</u> : (QID=25143)
Which of the following are considered CPE?
1. CSU/DSU
2. Telco switch
3. Router
4. Demarcation point
Correct answer(s): 1 3
Explanation: The CSU/DSU and router are considered CPE, Customer Premises Equipment.
[Implement and verify WAN links]
Question 42: (QID=2378)
What command do we use to determine if a Serial interface on a Cisco router is either DTE or DCE?
1. Show interface serialx
2. Show ip interface serialx
3. Show controllers serial x
4. Show role serial x
Correct answer(s): 3

1. DoS attack



Which of the following statements are true? [Select all that apply]

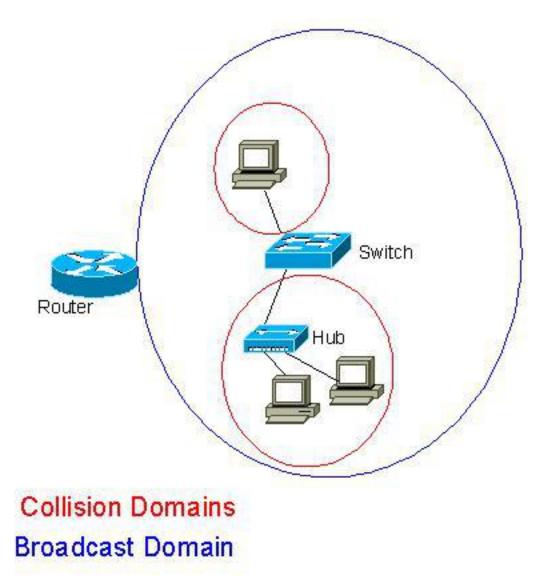
- 1. Switches control broadcast domains
- 2. Hubs control collision domains

<u>Question 45</u>: (QID=1479)

- 3. Routers control broadcast domains
- 4. Hubs control broadcast domains
- 5. Switches control collision domains

Correct answer(s): 35

Explanation: Routers are the border of a broadcast domain because they don't forward broadcasts by default. A switch can control a broadcast domain if VLANs are configured on the switch otherwise they divided the network in separate collision domains.



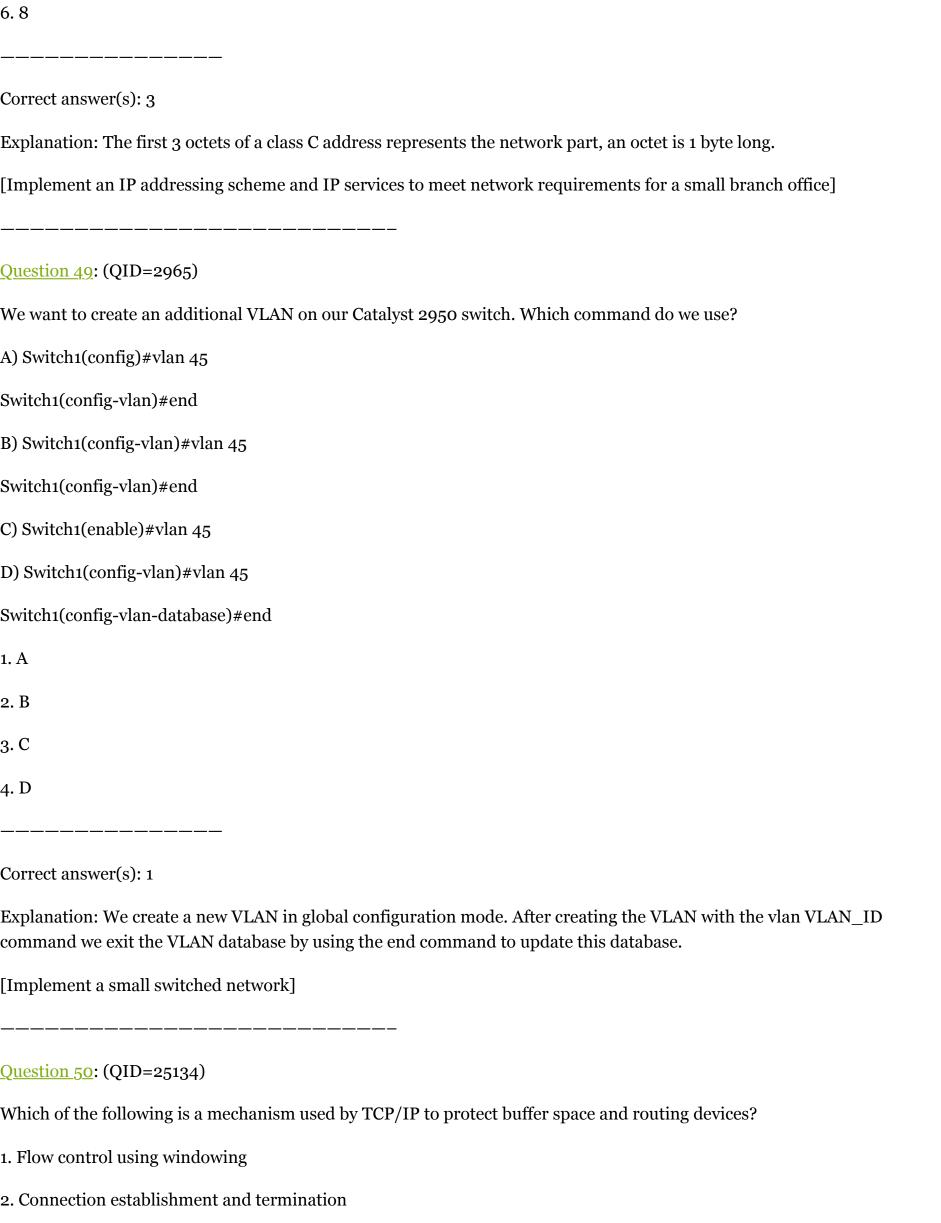
[Implement a small switched network]

<u>Question 46</u>: (QID=25138)

What is described by a self-propagating piece of software that replicates itself on the network and usually used in DoS

1. Virus
2. Trojan
3. Spyware
4. Worm
Correct answer(s): 4
Explanation: A worm is a program that self-propagates on the network without user intervention and is commonly used in DoS attacks.
[Identify security threats to a network and describe general methods to mitigate those threats]
Question 47: (QID=25122)
Which of the following terms describes an IP address that represents a host on an enterprise network?
1. Inside global
2. Inside local
3. Outside global
4. Outside local
Correct answer(s): 2
Explanation: The inside local address is the IP Address of a host that is local to the enterprise network.
[Implement and verify WAN links]
Question 48: (QID=25142)
How many bytes make up the network portion of a class C address?
1. 1
2. 2
3. 3
4.4
5. 6

attacks?



4. Multiplexing using ports			
Correct answer(s): 1			
Explanation: Flow control using windous transport layer of the OSI model.	owing is a mechanism used by TCP/IP to	protect buffer space and routing dev	rices in the
[Describe the operation of data networ	·ks]		
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50 Question Set In "CCNA"	(Blueprint) In "CCNA"	In "CCNA"	
III CCNA	III CCNA		
Make a Comment			

3. Error recovery

20 Responses to "CCENT Final Exam 1 - 50 Question Set"



This material has been very helpful through the course of my study for CISCO CERTIFICATION





Reply

It has excellent information. Explanation, and answers at the bottom made me easy to understant. Thanks to the editor.

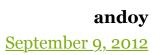




<u>Reply</u>

Question #34

The answer is wrong.





<u>Reply</u>

questio 33: the question and the answer doesn't match. The question sould rather include:

"You do not want R3 to send RIP updates"





Reply

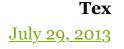
Hello to all, it's really a good for me to pay a quick visit this site, it consists of priceless Information.





Reply

Very helpful, thank you





<u>Reply</u>

Awesome!!

This was great!!

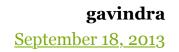
Really helpful!!





Reply

this was very helpful for my ccent certification gavindra sookhoo





<u>Reply</u>

https://kl2217.wordpress.com/2009/09/ccent-final-exam-1-50-question-set/[...]

<u>PART7 ICND1</u> <u>August 20, 2014</u>

ly

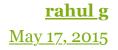
Hello to all, the contents existing at this site are really remarkable for people experience, well, keep up the nice work fellows.





<u>Reply</u>

Default AD for RIP is 110 not 90...it's confusing!





<u>Reply</u>

AD for RIP is 120 : AD 110 is for OSPF: AD 90 is for EIGRP \bigcirc





Reply

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<u>Reply</u>

Reply

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Certifications
April 3, 2016

lv

This practice exam was very good for pointing out the weak spots in my knowledge, and the explanations were clear, concise, and very helpful. Thank you so much for your time and effort in putting this together.





<u>Reply</u>

I have found by googling that website where to practice online ccent Practice the Cisco ICND1 v3.0(100-105) exam.

The questions from you test are from the same version?

http://www.e-smartsolution.co.uk/ccent

Thanks

J. smith
December 2, 2016



<u>Reply</u>

Bonjour

j' ai trouvé cet site de la version 3.0

préparation examen cisco ccent ICND1 v3.0(100-105)

http://www.e-smartsolution.co.uk/ccent

Salut!

<u>kımu</u> <u>December 6, 2016</u>



<u>Reply</u>

<u>Reply</u>

some of these questions are taken from moon...

waaaa112sabi October 23, 2017



<u>Reply</u>

Where's The Comment Form?

CSE 473 – Introduction to Computer Networks

Jon Turner

Final Exam Solution

12/18/2013

1. (10 points). What is the queueing delay at a network link with a link rate of 100 Mb/s, an arriving traffic rate of 9,000 packets per second an average packet length of 1250 bytes and a queue length of 500 packets?

The traffic intensity is 0.9, so the average number of packets in the queue is 9. Since the time needed to send one packet is 100 μ s, the average queueing delay is 900 μ s.

What is the delay if the arriving traffic rate increases to 15,000 packets per second?

In this case, the queue fills, so the average delay is about 50 ms.

Consider an audio application that sends audio data using RTP and UDP over an Ethernet network. How many overhead bytes does each packet include, assuming that the RTP header is 12 bytes? You may ignore the Ethernet preamble and flag.

The overhead is 18+20+8+12=58 *bytes.*

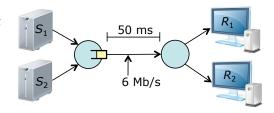
If audio is encoded at 32 Kb/s and we want or audio packets to be large enough so that the overhead is at most 20% of the total, how many audio bytes must each packet carry?

We need four times 58 or 232 bytes.

How long does it take to acquire the audio data needed to "fill" a packet?

At 32 Kb/s, we get four bytes of audio data per ms, so it takes 58 ms to fill the packet.

2. (15 points) The diagram at right shows two TCP senders at left and the corresponding receivers at right. The first sender uses TCP *Tahoe*, the second uses *Reno*. Assume that the MSS is 1 KB, that the one-way propagation delay for both connections is 50 ms and that the link joining the two routers has a bandwidth of 6 Mb/s. Let *cwnd*₁ and *cwnd*₂ be



the values of the senders' congestion windows. What is the smallest value of $cwnd_1 + cwnd_2$ for which the link joining the two routers stays busy all the time?

The RTT is 100 ms in this case, so the link rate is equivalent to 600 Kb per RTT or 75 KB. So, $cwnd_1+cwnd_2=75$ KB.

Assume that the link buffer overflows whenever $cwnd_1+cwnd_2\ge 150$ KB and that at time 0, $cwnd_1=30$ KB and $cwnd_2=120$ KB. Approximately, what are the values of $cwnd_1$ and $cwnd_2$ one RTT later? Also, what are the values of ssthresh for each of the two connections? Assume that all losses are detected by triple duplicate ACKs.

Since the first uses Tahoe and the second uses Reno, $cwnd_1=1$ KB and $cwnd_2=60$ KB, $ssthresh_1=15$ KB and $sshresh_2=60$ KB.

After 8 more RTTs, approximately what are the values of *cwnd*₁ and *cwnd*₂?

19 KB and 68 KB

Approximately, how many more RTTs before $cwnd_1+cwnd_2 \ge 150$ KB again? What is $cwnd_2-cwnd_1$ at this point?

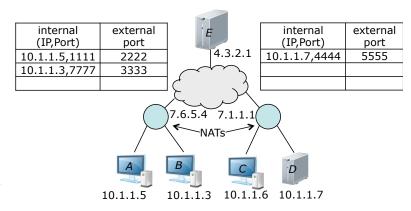
(150-87)/2=31.5 RTTs $cwnd_2-cwnd_1=49 KB$

Approximately, how many more RTTs pass before $cwnd_1+cwnd_2\geq 150$ KB and $cwnd_2-cwnd_1\leq 20$ KB?

The difference goes down by a little less than a factor of 2 after every cycle of the congestion control algorithm. So, 2 more cycles will be required to get the difference below 10 KB. So it will take about 2*(1+8+31.5)=81 RTTs.

3. (10 points). The figure at right shows two residential networks with routers that implement NAT. Suppose host *A* is connected to the web server at host *E*.

In the left-hand NAT table, add an entry that would allow *A* to communicate with *E*. You may choose any port numbers you like, but the internal port numbers should be different from the external port numbers.



Show the values of the address and port fields in the diagram below, for a typical packet sent by host *A*.

src adr	dest adr	src port	dest port
10.1.1.5	4.3.2.1	7777	80

Show the fields in the packet as it might appear when it reaches *E*.

_	src adr	dest adr	src port	dest port
	7.6.5.4	4.3.2.1	2222	80

Suppose the user in the right-hand network runs a game server on host *D* and invites her friends to join her game sessions. Add an entry to the right-hand table that would allow remote connections to the game server. Again, you may pick your own port numbers, but the internal and external port numbers should be different. Assume host *B* connects to the game server at *D*. Add an entry to the left-hand NAT table for this connection. Show the address and port fields for a typical packet leaving host *B*, the fields in the same packet as it passes through the public internet, and the fields in the packet that is delivered to *D*.

src adr	dest adr	src port	dest port
10.1.1.3	7.1.1.1	7777	5555
src adr	dest adr	src port	dest port
7.6.5.4	7.1.1.1	3333	5555
src adr	dest adr	src port	dest port
7.6.5.4	10.1.1.7	3333	4444

4. (15 points) Does SSL use public key encryption? If so, what does it use it for and how does it use it, in a typical situation?

Yes. Typically a server's public key is signed using a certificate authority's private key, so that a client can use its copy of the CA's public key to verify that that the key in the certificate really belongs to the server. The client then uses the server's public key to encrypt some secret information, which it sends to the server. This information is then used to generate additional keys.

Does SSL use symmetric encryption? If so, what does it use it for and how does it use it?

Yes. After the initial handshake, the client and server encrypt data using symmetric encryption.

How many different keys are created for a typical SSL session? What are they used for?

Four keys are created, two encryption keys (one for each direction) and two MAC keys (one for each direction.

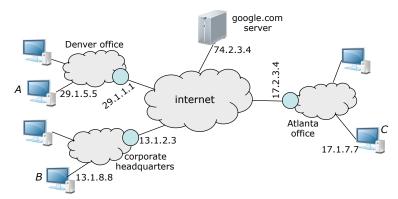
Is it necessary for a single SSL record to be sent entirely within one IP packet? Why or why not?

It is not. The TCP layer does not know about SSL records, it just sends bytes. So the IP packet boundaries may come in the middle of SSL records. This does no harm, as SSL is unaware of IP packet boundaries and simply operates on bytes.

Suppose an attacker learned the encryption key used by one endpoint of an SSL connection. Could the attacker use this to modify one of the records in the connection, without this being detected by the receiver? If so how, if not, why not?

No. While the attacker could encrypt the packet, without also knowing the MAC key, it could not compute a new MAC to go with the modified record. Consequently, the receiver would be able to detect the modification to the record.

5. (15 points) The diagram at right shows a corporate network with three sites connected by the internet. The company's n etwork administrator has configured the gateway routers at the three sites to use IPsec to encrypt all traffic going between the three sites. How many Security Associations are needed to allow the sites to communicate securely with each other?



Six SAs are needed, two for each pair of offices (one in each direction).

If host *A* sends a packet to host *C*, what are the source and destination address fields in the packet header as it passes through the public internet?

Source address is 29.1.1.1, destination address is 17.2.3.4.

If host *B* sends a packet to the Google server shown, what are the source and destination address fields in the packet header as it passes through the public internet?

Source address is 13.1.8.8, destination address is 74.2.3.4.

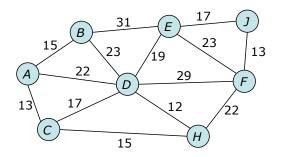
How does a gateway router decide if an outgoing packet should be encrypted using IPsec? How does it decide which Security Association to use, if it does require encryption?

It looks for a matching entry in its Security Policy Database (SPD), using the source and destination address fields in the packet as part of the lookup key. If there is a matching entry, the packet should be encrypted and the entry will include the Security Parameter Index (SPI) of the appropriate SA.

If the gateway router at Atlanta receives an IPsec packet from the public Internet, how does it determine which encryption key to use when decrypting the packet?

The packet will include the SPI of the Security Association in cleartext. The receiving router will lookup the entry for that SPI in its Security Association Database. This contains the encryption key for the SA.

6. (10 points) In the diagram at right, the nodes represent IP routers and the numbers on the links are OSPF link weights. If the network uses PIM with reverse-path f orwarding, which links are never used for multicast packets going to a host connected to router *J*?



The shortest path tree rooted at J includes the links EJ, FJ, BE, DE, CH, AD and FH. So the links that would never be used for multicast packets going to a host at J are: EF, DF, DH, BD, AB, AC, CD.

Suppose that router *C* is the PIM DR for some layer 2 subnet, and that a host *X* in that subnet sends an IGMP report for address 229.1.2.3. Assume that *J* is the RP for 229.1.2.3 and that no other routers are currently participating in 229.1.2.3. What does router *C* do at this point? Which routers eventually add forwarding state for 229.1.2.3 as a result of *C*'s action?

C would send a PIM Join packet to H. Eventually, routers C, H, F and J would add forwarding state for 229.1.2.3.

If a host connected to router *B* sent a packet to 229.1.2.3, which routers would the packet pass through in order to reach the host connected to the subnet at *C*? List them in the order in which the packet passes through them.

It would pass through routers B, E, J, F, H and C.

How would the previous answer change if router *C* issued a source-specific join on 229.1.2.3 for the host at router *B*.

This would cause packets from B to pass through routers B. A and C.

If the host connected to the subnet at *C* sends a packet to 229.1.2.3, which routers receive a copy? Assume that there has been no other activity on this multicast address, except for what has been described in the earlier parts.

Routers C, H, F and I would all receive copies. I would simply discard its copy.

7. (10 points) Consider an audio/video teleconferencing session with ten participants in the same switched layer 2 network. The application limits the number of simultaneous senders to two. If the RTP session bandwidth is 4 Mb/s, how much bandwidth can be used for RTCP packets, assuming the default behavior?

5% of the session bandwidth or 200 Kb/s.

Suppose that all participants send RTCP receiver reports and that at some point in time, two hosts are sending RTCP sender reports. How much bandwidth does each receiver have for its reports? How much does each sender have for its reports?

The RTCP bandwidth is split between senders and receivers with senders getting 25% of 200 Kb/s or 50 Kb/s. So each sender gets 25 Kb/s. Each receiver gets 15 Kb/s.

Suppose the number of participants in the conference increases to 20. In this case, how much bandwidth is available to each sender for its reports? How many for each receiver?

The senders still get 25 Kb/s each. The receivers get 7.5 Kb/s each.

When a participant in the conference receives an audio packet, how does it determine the actual time when the packet was sent, so that it can determine the appropriate playout delay?

The audio packet contains a timestamp that is based on the audio sample clock. This is essentially just just the value of an audio sample counter maintained by the source. To relate this to real-time, the receiver uses information it received from the last sender report issued by the audio source. This sender report relates the audio sample clock to the actual real-time.

8. (10 points) Consider a residential network that connects to the internet with a DSL link that has a download rate of 4 Mb/s. Assume that there are three UDP flows sharing the link and the remote hosts are sending at rates of 1 Mb/s, 2 Mb/s and 3 Mb/s. Assume that the ISP router has a link buffer that can hold 300 packets (assume all packets have the same length). For each flow, what fraction of the packets it sends are discarded?

They each lose about 33% of the packets they send.

For each flow, about how many packets does it have in the queue.

The first flow has about 50 packets in the queue, the second flow has about 100 and the third has 150.

Now, suppose the queue at the ISP router is replaced by three queues that can each hold 100 packets and that the queues are scheduled using weighted-fair queueing, where the weights are all 0.33. In this case, what fraction of packets are discarded from each flow?

In this case, the first flow loses 0, the second loses 25% and the third loses 50%.

How many packets does each flow have in the queue?

The first flow's queue is empty or close to empty. The other two each have about 100 packets in their queues.

Now, suppose the weights are 0.2 for the first flow, 0.6 for the second and 0.2 for the third. In this case, what fraction of packets are discarded from each flow?

In this case, the first two flows each lose 0, while the third loses 67%.

How many packets does each flow have in the queue?

The second flow's queue is empty or close to empty. The third has about 100 packets in its queue.

The first flow is a little tricky, as its allocated portion of the link bandwidth exactly matches its data rate. If the packets arrive with uniform spacing, the queue will remain close to empty, but if the times between packets are more random, significant queueing can occur. Indeed, there can even be some packet loss in this case. For a single flow however, it's more likely that the queue will never accumulate a large backlog.

9. (10 points) The diagram at right shows a WIFI network with an access point, *X* and three hosts, *A*, *B* and *C*. The large circles indicate the *coverage areas* of the three hosts. The coverage area for *X* is not shown, but you may assume that it includes all three hosts. Assume RTS/CTS are not used.

Suppose X is transmitting a packet at time 0 and finishes sending it at time 100 μ s. Also,

- A gets a packet to send at time 50 that takes 100 μ s to send and is assigned a backoff timer of 100.
- *B* gets a packet at time 70 that takes 200 µs and is assigned a backoff timer of 50.
- *C* gets a packet at time 90 that takes 150 µs and is assigned a backoff timer of 150.

For each of the three hosts, what time do they start sending their packets? You may ignore the inter-frame spacing and the time required for acks.

 \bigcirc

(C)

A starts sending at time 200 and finishes at 300 B starts sending at 150 and finishes at 350 C starts sending at 350 and finishes at 500

Of the three packets sent, which are successfully delivered on the first attempt?

Only the one from C is successfully delivered.

For each packet that is not successfully delivered on the first attempt, approximately when does the sending host learn that the packet was lost and must be sent again?

Hosts learn of lost packets from the absence of ACKs. Here, A would expect an ACK at 300 and so would learn of the lost packet when the ACK fails to arrive at time 300. Similarly, B would learn of its lost packet at time 350.

Now, suppose RTS/CTS is enabled. In this case, approximately when does each host send its data packet? You may assume that the time needed to send RTS, CTS and ACK packets is negligible.

A sends at 400.

B sends at 150.

C sends at 550.



University of Uppsala

Department of Computer Systems (DoCS)

Final Examination

IT3 (Datorsystem II : Networks)

Data Communication and Networks

INSTRUCTIONS TO CANDIDATES

- This is a FIVE (5) hour examination
- Answer all questions
- All questions to be answered in English
- Dictionaries are Permitted
- Marks total 180
- This exam contributes 85/100 points to the final mark in this subject.

A: (Short Answer Questions)

Question 1

Name two well known data transport protocols provided by the Internet Transport Layer.

Provide a brief description of each service and indicate what type of application might use that service.

[8]

SOLUTION

TCP and UDP.

TCP is a connection oriented data service that provides a reliable loss free end to end connection. Peer to peer communication between applications is often managed using TCP.

[4]

UDP is an unreliable datagram service which provides end to end packet delivery without recovery and retransmission for applications that might wish to implement their own packet management and error recovery protocols such as streaming applications and NFS. UDP is also often used in Client/Server applications where no special packet handling is required.

[4]

Describe the "token bucket" mechanism for congestion control.

With which other technique is token bucket usually combined to achieve complete flow control?

What problems in the simpler approach are addressed by using a "token bucket" mechanism?

[9]

SOLUTION

The token bucket approach to congestion tries to limit the impact of bursts of network packets on routers along the path, while retaining some flexibility in terms of allowing a brief increase in traffic in response to sudden demands/bursts.

The answer should include a picture indicating the burst response pattern that results from the use of token bucket, and show how that feeds into a leaky bucket in order to eliminate packet loss from the leaky bucket and provide overall loss free rate/flow control.

[4]

Token bucket is combined with leaky bucket, which buffers incomming packets and emits them in a constant rate stream.

[1]

The advantage of the combination is that token bucket provides some flexibility in responding to sudden requests for high traffic volume allowing an intial burst of traffic to saturate the link, and then enforcing a constant bandwidth utilisation after the initial free token pool has been exhausted.

[4]

For three marks each define the following general terms used when discussing protocols.

Peers

IDU

SDU

PDU

[12]

SOLUTION

- Peers active elements of a protocol (processes) at the same protocol stack layer level on different hosts.
- IDU Interface Data Unit, data to be passed accross an access point. Generally SDU plus control information.
- SDU Service Data Unit, data/information to be sent to a peer process.
- PDU Protocol Data Unit, packet data to be transmitted. May be a part of an SDU with a header. N-PDU's are the packet format exchanged between peers at layer N of the protocol stack.

Give a definition of a **Service** and a **Protocol**. Use these definitions or any other discussion to illustrate the fundamental difference between a **Service** and a **Protocol**.

[10]

SOLUTION

Defn: A service is an operation or facility provided by protocol layer N to the layer N+1 above it.

[3]

Defn: A protocol is an algorithm operating at protocol layer N that uses services provided by layer N-1 to implement a style of communication.

[3]

A service is a facility that can be used, such as **SendAPacket**. A protocol uses a number of services to manage the exchange of data. Thus a protocol may be compared to an algorithm, and the services to the instructions which are used to express that algorithm.

[4]

Give an overview of the distance vector method of updating routing table information. In particular, explain using an example how information about a node failure propagates using this algorithm.

What problem is encountered in deciding whether a host has become unreachable?

In what circumstances is it impossible to resolve this problem?

[16]

SOLUTION

Distance vector routing uses a distance metric, (round trip time, physical distance, number of network hops) to select the best current path to each reachable destination in the network.

To compute its routing table a host computes the value of the chosen metric with respect to all its neighbours, and also obtains the routing vectors of all its neighbours. It then computes the distance metric to a destination A for each (neighbour, destination) pair. The host then chooses the neighbour with the minimum distance value to be used to forward messages to the destination A. This process is repeated for all destinations in the network.

[8]

When a host fails in the network this information propagates very slowly, since each host updates its vector to be the minimum of its neighbours vector entries plus its delay to that neighbour. How do we decide that the delay vector entry has become high enough to register the host as unreachable? This is called the count to infinity problem.

[4]

The problem can be resolved if the distance metric is related to the number of hops to the destination. Then we can set infinity to be network diameter plus 1. If the metric is time based we need an estimate of the maximum acceptable delay. In this case it is possible that congestion will cause some hosts to be marked as unreachable from time to time. However, since host recovery propagates quite quickly this may not be a problem at least in relatively small networks.

[4]

Provide the formal definition of a Petri-Net and describe each of the variables/operators in the 5 entries in the 5-tuple.

[8]

SOLUTION

The formal definition of a Petri-Net C is a 5-tuple as follows.

$$C = (P, T, I, O, \mu)$$
 [1]

- [1] P The set of places in the net
- [1] T The set of transitions in the net
- [2] I The input mapping function, applied to a transition generates the set of input places to that transition.
- [2] O The output mapping function, applied to a transition generates the set of output places to that transition.
- [1] μ The initial marking, which describes the initial location of tokens.

Discuss the use of formal analysis techniques for protocols. Comment on why such techniques are used in analysing protocols, and give some examples of the types of problems that such an analysis can reveal.

[6]

SOLUTION

Formal analysis of protocols is an important part of establishing confidence in the correctness of proposed communications products. Formal analysis tools such as petri nets and finite state machines represent the logical structure of the protocol and allow designers to investigate the operation of the protocol in all possible situations.

State space expansion can identify failure modes, and deadlock situations that can arise if certain sequences of events occur. In addition, such analysis can reveal redundant states and thus be used to optimise the protocol. Safety properties can also be proposed and verified.

The use of some specification tools can assist in automatic verification using computer analysis which eliminates the possibility of human error, an important factor in many proof techniques.

It has been claimed that state machines are a natural tool for protocol analysis since protocol implementations can be derived directly from the state transition tables. Discuss this assertion and argue for or against the implementation of protocols using this approach.

[8]

SOLUTION

Since state/transition diagrams are directly related to table driven programming it can be argued that the type of specification and implementation flow directly from one another.

By deriving the implementation directly from the specification with little need for interpretation the possibility of erroneous implementations is decreased.

In addition it has been argued by Tannenbaum and others that state transition based programming is the natural way to implement protocols, and to visualise their operation. Thus the choice of state machines for the analysis and validation of protocols is both natural and effective. Especially since tools are available that are capable of generating implementation code directly from the specification.

Define the role of a **firewall** and draw a diagram that shows where a firewall should be positioned with relation to protecting a local network. In defining the role of a firewall, you should discuss the techniques that a firewall uses at different levels to prevent external attacks on the network and control traffic flow through the firewall.

[6]

SOLUTION

The solution will show a local area network structure and the firewall as the single interface point or gateway to the outside world.

[2]

The characterisation of firewall activity will probably touch on a majority of the following points.

- denying connection requests to known ports, such as ftp, sendmail on hosts within the local network.
- packet filtering based on keywords or encryption status.
- suspicious packet traffic patterns that represent denial of service or packet spoofing attacks on local hosts from external IP addresses.
- an exceptional answer might note that wireless networking often bypasses firewalls and represents a possible security hole.

[4]

Explain, using an example, how bit stuffing is used to preserve frame boundaries when transmitting binary data at the Data Link level of the protocol stack.

[8]

SOLUTION

Suppose we choose the bit pattern 011111110 to represent the frame boundary in a bit stream that we are transmitting. To prevent the occurence of this bit pattern in the payload (and thus incorrect identification of a frame boundary by the receiver we will modify the payload by inserting a 0 bit into the stream after every consecutive five 1 bits seen in the payload.

Unstuffing at the receiving end will remove these zero bits and restore the format of the original frame payload.

[4]

As an example consider the following payload bit stream.

The transmitted stream would be

preventing incorrect identification of a frame boundary in the highlighted section of the bit stream.

B: (Topic Questions)

Question 11

Describe the OSI seven layer model. Name each of the layers in the model and draw a diagram that shows the ordering of these layers. Write a paragraph describing the areas of function that each layer is responsible for.

Contrast the OSI model you have just described with the TCP/IP reference model. Using a diagram show the correspondence between relevant protocol layers in the two models.

Discuss the relative merits of each of these models in the context of modern computer networking.

[30]

SOLUTION

Presentation of the OSI diagram and list of the functions carried out in each layer.

[10]

The contrast with the TCP/IP model should note the corespondence between the OSI and TCP layers at the Network and Transport level. The discussion will also indicate that the Host_to_Network layer of TCP/IP corresponds pretty well to the Data Link and Physical layers of the OSI model. Session and Presentation layers are not present in TCP/IP, and really are not needed as has been shown by experience over the last few years.

[8]

A discussion of the merits of TCP/IP vs OSI as a reference model for networking. In terms of discussing relative merits students should touch on the following key ideas.

- Political influence on the OSI model, thus the OSI protocol stack represents a design philosophy created by committee
- OSI failed to gain wide acceptance in the industry and consequently is largely of academic interest.
- OSI provides a much better model of layering and clearly defines and separates protocols, interfaces and services. This allows for greater flexibility and allows the layers to be viewed and implemented independently of each other.
- TCP/IP describes a protocol, it is not really a standard.
- TCP/IP is in very wide use and is well understood.
- TCP/IP has proven very robust and well engineered.

[12]

Discuss the hierarchy and addressing issues surrounding the construction of large networks.

[10]

Comment on the current status of IPv4 in this context. Identify the major emerging problems for IPv4 and discuss how they are addressed in IPv6.

[20]

SOLUTION

A good solution to this question should discuss the problems associated with flat address structures in large networks.

In particular the following points:-

- length of addresses
- size of routing tables as a function of number of hosts.
- the concept of routing between network clusters on a backbone, and then routing within the destination cluster, so called "hierarchical routing".

In the context of IPv4 the discussion will center on the issues surrounding the packet header fields, and the current IP address space.

Use of class B networks for small numbers of hosts and the resulting waste in the address space should be discussed, and the the use of CIDR to resolve some of these problems in the near future should be described.

Consideration of IPv6 should touch on the arguments associated with selecting the new address format, and the elimination of the checksum field and other changes to the packet header.

Describe the physical construction characteristics of the following transmission media. If there is more than one type of each media then present the choices of type, and identify the differences between them.

- Twisted pair
- Coaxial cable
- Optical fibre cable

[9]

Possible choices for data communication infrastructure for long distance data transfer include microwave, optical fibre and radio transmission.

Discuss the advantages and disadvantages of each of these media, in terms of cost, political impact, ease of installation and adaptability in providing a range of services to clients.

[15]

What problem with data transmission in broadband coaxial cable networks is addressed using frequency splitting? Name two types of frequency splitting strategy, and identify how they are different.

[6]

SOLUTION

The discussion of communication media types should be structured similarly to the following.

• Twisted pair is the common telecommunications wiring standard for office and domestic buildings throughout the 1970's and 1980's. In the late 1980's UTP, (Uninsulated Twisted Pair)