

TDTS06 – Computer Networks

Final Examination: 14:00-18:00, Friday, Oct. 28, 2016

Time: 240 minutes

Total Marks: 40

Grade Requirements: three (20/40); four (28/40); and five (36/40).

Assistance: None (closed book, closed notes, and no electronics)

Instructor: Niklas Carlsson

**Instructions:**

- Read all instructions carefully (including these)!!!! Some questions have multiple tasks/parts. Please make sure to address *all* of these.
- The total possible marks granted for each question are given in parentheses. The entire test will be graded out of 40. This gives you 10 marks per hour, or six minutes per mark, plan your time accordingly.
- This examination consists of a total of 9+1=10 questions. Check to ensure that this exam is complete.
- When applicable, please explain how you derived your answers. Your final answers should be clearly stated.
- Write answers legibly; no marks will be given for answers that cannot be read easily.
- Where a discourse or discussion is called for, be concise and precise.
- If necessary, state any assumptions you made in answering a question. However, remember to read the instructions for each question carefully and answer the questions as precisely as possible. Solving the *wrong* question may result in deductions! It is better to solve the *right* question incorrectly, than the *wrong* question correctly.
- Please write your AID number, exam code, page numbers (even if the questions indicate numbers as well), etc. at the top/header of each page. (This ensures that marks always can be accredited to the correct individual, while ensuring that the exam is anonymous.)
- Please answer in English to largest possible extent, and try to use Swedish or "Swenglish" only as needed to support your answers.
- If needed, feel free to bring a dictionary from an official publisher. Hardcopy, not electronic!! Also, your dictionary is not allowed to contain any notes; only the printed text by the publisher.
- Good luck with the exam.



**1) Question: Forwarding (6)**

Show, illustrate, and explain the path of (i) the first *SYN packet* and (ii) the *third HTTP response* between a Web client (browser) and a Web server. You can make the following assumptions:

- The client is located in Sweden and the server in Australia.
- Koalas are cute!
- The client machine uses Ethernet, has a single interface with a MAC address AA:AA:AA:AA:AA:AA.
- The GET requests of interest here are for a website (*www.aa.com/index.html*) with two embedded images. You can assume that the images are hosted on the same website but are located under a directory *images/* and are named *imageAA.gif* and *imageBB.gif*. You can also assume that *imageAA.gif* is requested before *imageBB.gif*.
- The client has obtained a dynamic IP address 123.123.65.123 from a DHCP server, which is running on the closest gateway router.
- The client uses a local DNS server with IP address 123.123.68.44 and MAC address EE.EE.EE.EE.EE.EE.
- The MAC and IP addresses of the Web server are DD:DD:DD:DD:DD:DD and 200.82.111.111. Similar to the client, the server has a single interface.
- The gateway router closest to the client has four interfaces. The first is the interface closest to the client and has MAC and IP addresses B1:B1:B1:B1:B1:B1 and 123.123.65.1. The second interface has MAC and IP addresses B2:B2:B2:B2:B2:B2 and 123.123.66.2. The third interface has MAC and IP addresses B3:B3:B3:B3:B3:B3 and 123.123.67.3. Finally, the fourth interface has MAC and IP addresses B4:B4:B4:B4:B4:B4 and 123.123.68.4.
- The gateway router closest to the server has four interfaces. The first is the interface closest to the server and has MAC and IP addresses C1:C1:C1:C1:C1:C1 and 200.82.111.1. The second interface has MAC and IP addresses C2:C2:C2:C2:C2:C2 and 200.82.112.2. The third interface has MAC and IP addresses C3:C3:C3:C3:C3:C3 and 200.82.113.3. Finally, the fourth interface has MAC and IP addresses C4:C4:C4:C4:C4:C4 and 200.82.114.4.
- The forwarding table at the gateway router closest to the client has many entries. However, for this question, the four most closely related entries states: 200.82.96.0/19 over interface 2, 200.82.64.0/20 over interface 2, 200.82.96.0/20 over interface 3, and 200.82.104.0/21 over interface 4.
- The forwarding table at the gateway router closest to the server has many entries. However, for this question, the four most closely related entries states: 123.123.192.0/18 over interface 2, 123.123.0.0/18 over interface 2, 123.123.64.0/18 over interface 3, and 123.123.0.0/16 over interface 4.

For this question you should also draw a picture of the topology, clearly indicate the path taken by the packet in this topology, and clearly state any assumptions you make about the topology (including parts of the networks not explained above) or anything else needed to solve the question. As with all your answers it is important that you also explain how you derived your answer. For example, why was the packet taking this particular route and not some other route?



**2) Question: Encapsulation (4)**

Consider the same scenario as above (Q1). Please use figures to show and illustrate the link-layer frame and encapsulated information for:

1. the first *DNS response* message when it reaches the *incoming* network interface card (link layer) of the *client*; and
2. the second *HTTP request* message when it reaches the (on the gateway) *incoming* network interface card (link layer) of the *gateway closest to the client*.

You do not have to show all the details of the different headers; however, you should provide a figure that clearly (i) specify what protocols the different headers are associated with, and (ii) provide the address information associated with the source and destination fields for each of the different headers contained within these two frames (i.e., address information at all layers, including the application layer).

**3) Question: TCP slow start (4)**

Consider the download of the above file *index.html*, located on the server hosting *www.aa.com*. Assume that the client and server are using non-persistent HTTP/1.0, and that there is a 100ms round trip time (RTT) between them. Please draw a figure and explain the entire communication sequence associated with the file download, including the TCP handshake and connection teardown. You can assume that the payload is 22 packets, that each packet can be sent in 1ms, and that the ninth (13<sup>th</sup>) payload packet is lost. You can also assume that the TCP version is implementing fast retransmit and fast recovery, that the initially *ssthresh* value is 4, and that the timeout period is (approximately) constant at 200ms. You should include a figure that clearly show when each packet is sent and received.

**4) Question: HTTP and replication (6)**

Performance and personalized service are important aspects of building good Web services. Please draw a picture illustrating the communication sequence when the client above downloads a smaller version of the above file *www.aa.com/index.html*, which in this question is only 4KB, but the webpage instead include a total of six embedded images. The main document and three of the embedded images can be found on the original Web server *www.aa.com* and the final three embedded images can be found on the server *www.bb.com*. You can assume that no proxy cache is used and the client communicates directly with the server *www.aa.com* using *pipelined HTTP* and directly with the server *www.bb.com* using *persistent HTTP but without pipelining*. Your picture should illustrate the client, all involved Web servers, and the communication sequence. Your picture should also clearly show connection establishment and teardown messages, as well as any other messages needed for the file transfer, so as to capture the relative timing of the different messages taking place over the different connections. You can assume that the HTML page and each of the images are 4KB each, the MSS is 1.5KB, and both webservers have the same RTT and no losses takes place during the example scenario. Also, how many packets are being sent between each of the two servers and the client? Which connections are likely to terminate last?



**5) Question: Distance vector routing (4)**

Consider a node A with neighbors B, C, and D. Node A currently has the distance table below. (Note that the network currently is not in a very good shape, and a few routing table updates will be needed to get the network back into shape.) Assume that it receives an updated distance vector from neighbor B which looks as follows  $[\infty, 0, 1, 3, 5, 3, 5]$  and an updated distance vector from neighbor C which looks as follows  $[\infty, 2, 0, 7, 2, 2, 4]$ . First, update the table below, including A's own distance vector. Second, assume that poison reverse is implemented, and show and explain exactly what information the node sends to each its neighbors (after the table has been updated).

Destination	Costs			
	A (via)	B	C	D
A	0 (A)	$\infty$	$\infty$	$\infty$
B	1 (B)	0	4	5
C	1 (C)	2	0	7
D	1 (D)	7	4	0
E	? (?)	6	8	7
F	? (?)	2	11	3
G	? (?)	10	5	1

**6) Question: IP fragmentation (4)**

Consider a 4,000 byte IPv4 datagram which traverses 10 links on its path from A to B. Assume that links 5, 6 and 8 have a maximum transmission unit (MTU) of 1,500 bytes and the other links have an MTU of 10,000 bytes.

- (a) Please use a figure and explain when fragmentations occurs, and into how many fragments the datagram is fragmented/defragmented as it traverses each link of the forwarding path from A to B.
- (b) Please use a figure and explain how this scenario will change if all routers on the path used IPv6. You can assume that the MTUs are the same as for the IPv4 case.

**7) Question: TCP fairness (4)**

Use figures and examples to illustrate why TCP's additive increase and multiplicative decrease (AIMD) mechanism provides some stability and fairness. Please consider two scenarios. In both scenarios there are two competing connections with the same shared bottleneck. In the first scenario, both connections should have the same RTT but start with different window size. In the second scenario, one of the connections should have twice the RTT as the other connection. Please show how stability and fairness is achieved over time in these two scenarios.





**8) Question: BGP routing (4)**

Consider the following information visible to a Border Gateway Protocol (BGP) router in an Autonomous System (AS) A.

- There are four neighboring ASes: B, C, D, and E.
- AS A's internal path costs to the closest gateway routers of each of these three ASes are 2, 3, 4, and 5, respectively.
- A is a customer of B and E; A is peering with C and D.
- For prefix 11.22.0.0/20 AS A sees the following route announcements: B Y Z; C F Z; D Q P Z; and E R Z.
- The last web request that a client located within AS A did to a server located in AS B, C, D, E had a RTT of 25ms, 35ms, 45ms, and 55ms, respectively.
- The last traceroute that somebody in AS B performed to a server or router in AS Z required 13 router hops. The corresponding numbers for traceroute queries between ASes C-Z, D-Z, and E-Z are 12, 15, and 14, respectively.

Please (i) draw a picture of the topology, as seen by AS A, (ii) clearly *list and order* the information that is being used when determining this routing table entry, (iii) clearly *rank the four path alternatives* (over B, C, D, and E), and (iv) show and carefully explain how the particular entry for the above prefix would look in the router's forwarding table. Note that the order should clearly reflect the priority (and order) used in making the routing decision (and similar routing decisions).

**9) Question: HTTP-based Adaptive Streaming (4)**

In mobile environments, the network conditions experienced by a client may vary significantly over the duration of a session. In this context, you should first explain what the main advantages of using HTTP-based adaptive streaming (HAS, used by Netflix, YouTube, DASH.js, Silverlight, Apple HLS, most TV stations, for example) relative to using non-adaptive HTTP-based streaming. Your answer should provide concrete examples (preferably using a figure of buffer conditions, network conditions, and playback quality, for example) of when and why HAS performs better in your example scenarios. Second, in the context of cache performance, please explain the potential advantages of (i) splitting a file into multiple chunks and (ii) using HTTP.

**10) Bonus Questions: Handovers and indirect routing in cellular (4)**

Explain and illustrate how handovers and indirect routing is used in the context of cell-phone networks (such as GSM) with a mobile client, so as to ensure that a user obtains seamless service. Please draw one or more figures that illustrate what happens with the routing of the network traffic as a mobile user that is away from its home network moves along a road, for example. Consider a mobility scenario involving many base stations and mobile switching centers.

Good luck!!

