Midterm

- [1] 1)
- [2] 1), 2), 3)
- [3] 1)
- $[4] \ 2)$
- [5] 1), 3)
- [6] 2)
- $[7] \ 4)$
- $[8] \ 2)$

[Computer Networks in general]

- [1] Autonomous means that the system works as it is without any external interaction. For instance, a meeting with a closed room is an example of autonomous system.
- [2] Protocols define how two different entity communicate. For instance, an application layer protocal defines four properties: the types of messages, the syntax and semantics of message fields and their alignments, and rules for sending and receiving the messages. Without protocols, computer devices cannot communicate each other due to lack of knowledges of how each other's messages are made.

[3]

- 1) 5 users.
- 2) 0.0729 (7.29%)
- [4] [I'll assume the broken characters are R]
 - 1) Assuming no other traffic in the network, the throughput for the file transfer is the minimum of the rates of links, which is $\min(R_1, R_2, R_3) = 200 \,\text{kbps}$.

2)
$$\frac{2\,\mathrm{MB}}{200\,\mathrm{kbps}} = \frac{16\times10^6\,\mathrm{bits}}{2\times10^5\,\mathrm{bps}} = 80\,\mathrm{seconds}.$$

3) The throughput is now $\min(R_1, R_2, R_3) = 500 \,\mathrm{kbps}$, and hence the transfer delay is roughly

$$\frac{2\,\mathrm{MB}}{500\,\mathrm{kbps}} = \frac{16\times10^6\,\mathrm{bits}}{5\times10^5\,\mathrm{bps}} = 32\,\mathrm{seconds}.$$

[5] Internet protocol stack is as follows: Application layer (HTTP) / transport layer (TCP or UDP) / network layer (IP) / link layer (ethernet, WiFi, etc.) / physical layer (copper wire, optical fiber, etc.)

The Internet's architecture got its hourglass shape as every network communication in the Internet should pass through IP.

[Application layer]

[1] The server differentiates the applications by the port numbers assigned to those applications.

[2]

- 1. User's PC requests to the local DNS server.
- 2. If the local DNS server does not have any cache for the requested domain name "google.com", request to the 'root server'.
- 3. The 'root server' returns the IP address of the '.com server.'
- 4. The local DNS server requests to the '.com server.'
- 5. The '.com server' returns the IP address of the 'google authoritative server.'
- 6. The local DNS server requests to the 'google authoritative server.'
- 7. The 'google authoritative server' resolves the 'google.com' and tell the IP address of it to the local DNS server.
- 8. The local DNS server sends the IP address of the 'google.com' to the user's PC.
- [3] No. the code s.recv(length-16) may return a message with less than length-16 bytes. Therefore, we need to repeatedly receive the pakeet from the socket s in order to ensure that the message is completely received.
- [4] I think the problem is fairly broken. No formulae are shown.
 - 1) By visiting n DNS servers, we need $n \cdot RTT$, and by requesting the page and the object, we need additional $8 \cdot RTT$. So in total (n+8)RTT is needed.

- 2) By visiting n DNS servers, we need $n \cdot RTT$, and by requesting the page and the object, we need additional $2 \cdot RTT$ by using parallel connections. So in total (n+2)RTT is needed.
- 3) By visiting n DNS servers, we need $n \cdot RTT$, and by requesting the page and the object, we need additional RTT by using persistent connection. So in total (n+1)RTT is needed.
- [5] Even if protocols provide various reliability, for instance, when a wrong message is received from the other, it may occur a panic. Thus, the end device should check the reliability once more.

[Transport Layer]

- [1] Suppose the message comes frequently from the other device. When we use NAK only, packet losses will be detected by missing sequence numbers quickly and corrupted packets will be detected by the NAK set by the other. So there is no problem. Also, by omitting ACK, we can get a performance gain. However, When the message comes infrequently, packet losses will not be detected until the we receive the next message. So we cannot differentiate between the packet loss and no message coming from the other device. It makes a problem. In summary, when the message comes frequently, NAK-only protocal have pros on performance; but when the message comes infrequently, NAK-only protocal have cons on reliability.
- [2] 1. Sender: Sends pkt0 and activate wait ACK0 state.
- 2. Receiver: Receives pkt0, sends ACK0 (with checksum), and activate 'wait for 1 from below' state.
- 3. Sender: Receives ACK0, sends pkt1, and activate wait ACK1 state.
- 4. Receiver: Receives pkt1, sends ACK1 (with checksum), and activate 'wait for 0 from below' state. But ACK1 is delayed.
- 5. Sender: Did not receive ACK1 and timed out, so resend pkt1 and remain in wait ACK1 state.
- 6. Receiver: receives pkt1 so resend ACK1 and remain in 'wait for 0 from below' state. 5. Sender: receives ACK1 and this solves the problem.

[3]

1) Go-Back-N: when the sender did not receive the ACK for a packet and it timed out, the sender resends all the packets in the window from the oldest non-acknowledged packet. It may retransmit an acknowledged packet. When the bits in sequence numbers are k, the maximum window size is $2^k - 1$.

Selective Repeat: the sender only resends non-acknowledged packets

in the window when the sender did not receive the ACK. It avoids unnecessary resending. When the bits in sequence numbers are k, the maximum window size is 2^{k-1} .

- 2) a. GBN: packets $W/2, W/2+1, \cdots, 3W/2-1$ will be retransmitted; SR: packet W/2 will be retransmitted.
 - b. GBN: At the time $t = T \cdot W/2$, the timer for packet W/2 starts. After the timeout of that timer, i.e., at $t = T \cdot W/2 + TO$, packets $W/2, \ldots, W-1$ will be retransmitted. The transmission delay is $T \cdot W/2$ and the propagation delay will be RTT/2. Thus, in total, we need a time of $T \cdot W + TO + RTT/2$.

SR: At the time $t = T \cdot W/2$, the timer for packet W/2 starts. After the timeout of that timer, i.e., at $t = T \cdot W/2 + TO$, packet W/2 will be retransmitted. The transmission delay is T and the propagation delay will be RTT/2. Thus, in total, we need a time of $T \cdot (W/2 + 1) + TO + RTT/2$.

c. SR, since it has smaller delay when a single packet is lost.

[4]

- 1) 1 to 6, 23 to 26.
- 2) 6 to 16, 17 to 22.
- 3) A triple duplicate ACK.
- 4) A timeout.
- 5) 24.
- 6) 12.
- 7) At 6th transmission round.
- [5] Enough thoughput and enough burst tolerance for long flows, short delay enough for short flows.

[Discussion & Suggestion]

Question 4 in application layer section seems corrupted.

[4] Suppose within your Web browser you click on a link to obtain a Web page. The IP address for the associated URL is not cached in your local host, so a DNS lookup is necessary to obtain the IP address. Suppose that n DNS servers are visited before your host receives the IP address from DNS; the successive visits incur an of,...,. Further suppose that the Web page associated with the link contains exactly one object, consisting of a small amount of HTML text. Let, denote the between the local host and the server containing the object. Suppose the HTML file references eight very small objects on the same server. Assuming zero transmission time of the object, how much time elapses from when the client clicks on the link until the client receives the object with each of the following cases? [6 points]

- 1) Non-persistent HTTP with no parallel TCP connections
- 2) Non-persistent HTTP with the browser configured for 5 parallel connections
- Persistent HTTP
- [5] In the presence of various threats in a file transfer application, as mentioned in 'end-to-

Thanks for your nice lectures.