

# CSCI 415 Data Communication Networks

## Homework 5

### Due 03/26/08

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#### **Problem 1: File Transfer**

Consider a simple UDP-based protocol for requesting files (based somewhat loosely on the Trivial File Transfer Protocol, TFTP). The client sends an initial request, and the server answers (if the file can be sent) with the first data packet. Client and server continue with a stop and wait transmission mechanism.

- (a) Describe a scenario by which a client might request one file but get another; you may allow the client application to exit abruptly and be restarted with the same port.
- (b) Propose a change in the protocol that will make this situation much less likely.

#### **Problem 2: Internet checksum**

In ones complement arithmetic, a negative integer  $-x$  is represented as the complement of  $x$ , that is each bit of  $x$  is inverted (that's 1111111111111111  $-x$ , hence the name). Therefore, 5 is 000000000000101, and -5 is 111111111111010. Similarly, 3 is 000000000000011 and -3 is 111111111111100. When adding numbers in ones complement, a carryout from the most significant bit must be added to the result (unlike in twos complement that is used in most machines). Therefore, if we add 111111111111010 and 111111111111100 ignoring the carry we get 111111111110110. In ones complement arithmetic, the fact that this operation caused a carry from the most significant bit causes us to increment the result, giving 111111111110111, which is the ones complement representation of -8 (obtained by inverting the bits), as we would expect.

The Internet checksum algorithm is the following: data is checksummed as a sequence of 16 bit integers. They are added together using a 16 bit ones complement arithmetic and the checksum is obtained as the ones complement of the result. The receiver adds all 16 bit words received including the checksum and, therefore, must obtain 111111111111111 (the ones complement of 0000000000000000) which is also a zero in ones complement).

This problem is designed to help you appreciate the Internet checksum algorithm. We start by asking the following questions:

- Beside the ease of implementation in software, how do we justify the use of a checksum?
- If the use of a checksum is justified, why ones complement arithmetic?

(a) Why a checksum (incremental updates)? Assume a router needs to change a word in the header from  $x$  to  $y$  before forwarding the packet, e.g. decrementing the TTL (time to live) of the packet. Argue that the new checksum can be obtained by simply adding  $x + \bar{y}$  to the previous checksum, where  $\bar{y}$  is the ones complement of  $y$ .

(b) Why ones complement (endian independent)? Little endian computers store numbers with the least significant byte first (Intel processors for example). Big endian computers put the most significant byte first (IBM mainframes for example). Show that adding 16 bit numbers in ones complement arithmetic is endian independent. More specifically, if  $AB+CD=EF$ , then  $BA + DC = FE$  (this is not necessarily true if ones complement arithmetic is not used), where  $A$ ,  $B$ ,  $C$ ,  $D$ ,  $E$ , and  $F$  are 8 bit numbers and, for instance,  $AB$  is a 16 bit number with  $A$  being the most significant byte and  $B$  being the least significant byte.

### **Problem 3: Wait before closing the connection**

Read about the netstat Unix/Windows utility (e.g. the man page for netstat on Unix). Use the netstat utility to see the state of your local TCP connections. Find out how long closing connections spend in the waiting state. For instance, you can use your server/client program from the second homework and observe how long it takes the client to close the connection. What happens if another client contacts the server while the previous client is in the wait state (observe the port numbers).

### **Problem 4: Designing a protocol header**

You are hired to design a reliable byte-stream protocol that uses a sliding window (like TCP). This protocol will run over a 1 Gbps network. The RTT of the network is 140 ms, and the maximum segment life is 60 seconds. How many bits would you include in the Advertised Window and Sequence Number fields of your protocol header?

### **Problem 5: Sequence numbers and fooling the server**

If server  $A$  accepts a TCP connection from client  $B$ , then during the three-way handshake  $A$  sent its initial sequence number to  $B$  and received an acknowledgment from it. Therefore, another client  $C$  can pretend to be  $B$  in the following way:

- $C$  sends a SYN packet to open TCP connection to  $A$  pretending to be  $B$
- $A$  sends its SYN+ACK packet with its initial sequence number as part of the three-way handshake to  $B$ , and waits for the acknowledgement
- $B$  ignores, i.e. does not respond to  $A$ 's SYN+ACK packet

- $C$  sends an acknowledgement to  $A$  pretending to be  $B$

One would argue, however, that  $C$  cannot properly acknowledge  $A$  because it does not receive  $A$ 's initial sequence number.

The algorithm for choosing the initial sequence number gives unrelated hosts a fair chance of guessing it. Specifically,  $A$  selects the initial sequence number based on a clock value at the time of connection. RFC 793 specifies that this clock value be incremented every  $4 \mu s$ ; common Berkeley implementations once simplified this to incrementing by 250000 once per second.

(a) Given this simplified increment-once-per-second implementation, explain how  $C$  could masquerade as  $B$  in at least the opening of a TCP connection.

(b) Assuming real RTT can be estimated to within 40 ms, about how many tries would you expect it to take to implement the strategy of part (a) with the unsimplified "increment every  $4 \mu s$  TCP implementation?"