

## Chapter 2 Review Questions

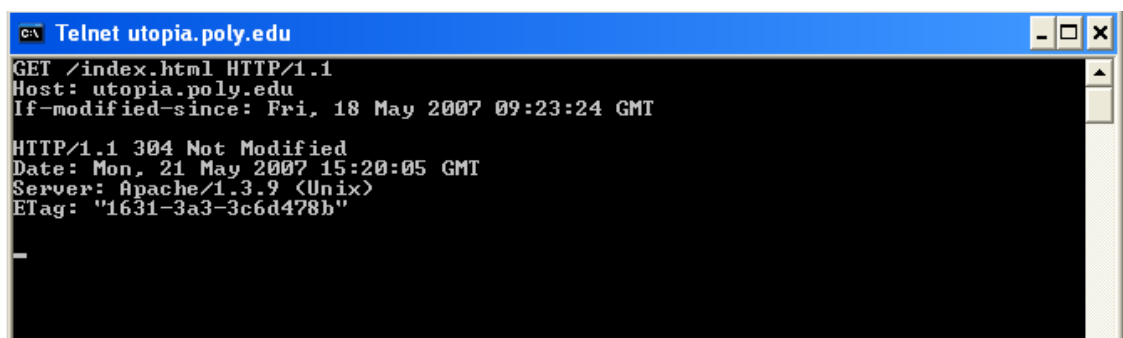
1. The Web: HTTP; file transfer: FTP; remote login: Telnet; Network News: NNTP; e-mail: SMTP.
2. Network architecture refers to the organization of the communication process into layers (e.g., the five-layer Internet architecture). Application architecture, on the other hand, is designed by an application developer and dictates the broad structure of the application (e.g., client-server or P2P)
3. The process which initiates the communication is the client; the process that waits to be contacted is the server.
4. No. As stated in the text, all communication sessions have a client side and a server side. In a P2P file-sharing application, the peer that is receiving a file is typically the client and the peer that is sending the file is typically the server.
5. The IP address of the destination host and the port number of the destination socket.
6. You would use UDP. With UDP, the transaction can be completed in one roundtrip time (RTT) - the client sends the transaction request into a UDP socket, and the server sends the reply back to the client's UDP socket. With TCP, a minimum of two RTTs are needed - one to set-up the TCP connection, and another for the client to send the request, and for the server to send back the reply.
7. There are no good examples of an application that requires no data loss and timing. If you know of one, send an e-mail to the authors.
8. a) Reliable data transfer  
TCP provides a reliable byte-stream between client and server but UDP does not.  
  
b) A guarantee that a certain value for throughput will be maintained  
Neither  
  
c) A guarantee that data will be delivered within a specified amount of time  
Neither  
  
d) Security  
Neither
9. SSL operates at the application layer. The SSL socket takes unencrypted data from the application layer, encrypts it and then passes it to the TCP socket. If the

application developer wants TCP to be enhanced with SSL, she has to include the SSL code in the application.

10. A protocol uses handshaking if the two communicating entities first exchange control packets before sending data to each other. SMTP uses handshaking at the application layer whereas HTTP does not.
11. The applications associated with those protocols require that all application data be received in the correct order and without gaps. TCP provides this service whereas UDP does not.
12. When the user first visits the site, the site returns a cookie number. This cookie number is stored on the user's host and is managed by the browser. During each subsequent visit (and purchase), the browser sends the cookie number back to the site. Thus the site knows when this user (more precisely, this browser) is visiting the site.
13. Web caching can bring the desired content "closer" to the user, perhaps to the same LAN to which the user's host is connected. Web caching can reduce the delay for all objects, even objects that are not cached, since caching reduces the traffic on links.
14. Issued the following command (in Windows command prompt) followed by the HTTP GET message to the "utopia.poly.edu" web server:

```
> telnet utopia.poly.edu 80
```

Since the index.html page in this web server was not modified since Fri, 18 May 2007 09:23:34 GMT, the following output was displayed when the above commands were issued on Sat, 19 May 2007. Note that the first 4 lines are the GET message and header lines input by the user and the next 4 lines (starting from HTTP/1.1 304 Not Modified) is the response from the web server.



```
C:\> Telnet utopia.poly.edu
GET /index.html HTTP/1.1
Host: utopia.poly.edu
If-modified-since: Fri, 18 May 2007 09:23:24 GMT

HTTP/1.1 304 Not Modified
Date: Mon, 21 May 2007 15:20:05 GMT
Server: Apache/1.3.9 (Unix)
ETag: "1631-3a3-3c6d478b"
```

15. FTP uses two parallel TCP connections, one connection for sending control information (such as a request to transfer a file) and another connection for actually transferring the file. Because the control information is not sent over the

same connection that the file is sent over, FTP sends control information out of band.

16. Message is sent from Alice's host to her mail server over HTTP. Alice's mail server then sends the message to Bob's mail server over SMTP. Bob then transfers the message from his mail server to his host over POP3.

17.

```
Received:      from 65.54.246.203 (EHLO bay0-omc3-s3.bay0.hotmail.com)
               (65.54.246.203) by mta419.mail.mud.yahoo.com with SMTP; Sat, 19
               May 2007 16:53:51 -0700
Received:      from hotmail.com ([65.55.135.106]) by bay0-omc3-s3.bay0.hotmail.com
               with Microsoft SMTPSVC(6.0.3790.2668); Sat, 19 May 2007 16:52:42 -
               0700
Received:      from mail pickup service by hotmail.com with Microsoft SMTPSVC; Sat,
               19 May 2007 16:52:41 -0700
Message-ID:    <BAY130-F26D9E35BF59E0D18A819AFB9310@phx.gbl>
Received:      from 65.55.135.123 by by130fd.bay130.hotmail.msn.com with HTTP;
               Sat, 19 May 2007 23:52:36 GMT
From:          "prithula dhungel" <prithuladhungel@hotmail.com>
To:            prithula@yahoo.com
Bcc:
Subject:       Test mail
Date:          Sat, 19 May 2007 23:52:36 +0000
Mime-Version:  1.0
Content-Type:  Text/html; format=flowed
Return-Path:   prithuladhungel@hotmail.com
```

**Figure: A sample mail message header**

*Received:* This header field indicates the sequence in which the SMTP servers send and receive the mail message including the respective timestamps.

In this example there are 4 "Received:" header lines. This means the mail message passed through 5 different SMTP servers before being delivered to the receiver's mail box. The last (forth) "Received:" header indicates the mail message flow from the SMTP server of the sender to the second SMTP server in the chain of servers. The sender's SMTP server is at address 65.55.135.123 and the second SMTP server in the chain is by130fd.bay130.hotmail.msn.com.

The third "Received:" header indicates the mail message flow from the second SMTP server in the chain to the third server, and so on.

Finally, the first "Received:" header indicates the flow of the mail message from the forth SMTP server to the last SMTP server (i.e. the receiver's mail server) in the chain.

*Message-id:* The message has been given this number BAY130-F26D9E35BF59E0D18A819AFB9310@phx.gbl (by bay0-omc3-s3.bay0.hotmail.com. Message-id is a unique string assigned by the mail system when the message is first created.

*From:* This indicates the email address of the sender of the mail. In the given example, the sender is “prithuladhungel@hotmail.com”

*To:* This field indicates the email address of the receiver of the mail. In the example, the receiver is “prithula@yahoo.com”

*Subject:* This gives the subject of the mail (if any specified by the sender). In the example, the subject specified by the sender is “Test mail”

*Date:* The date and time when the mail was sent by the sender. In the example, the sender sent the mail on 19<sup>th</sup> May 2007, at time 23:52:36 GMT.

*Mime-version:* MIME version used for the mail. In the example, it is 1.0.

*Content-type:* The type of content in the body of the mail message. In the example, it is “text/html”.

*Return-Path:* This specifies the email address to which the mail will be sent if the receiver of this mail wants to reply to the sender. This is also used by the sender’s mail server for bouncing back undeliverable mail messages of mailer-daemon error messages. In the example, the return path is “prithuladhungel@hotmail.com”.

18. With download and delete, after a user retrieves its messages from a POP server, the messages are deleted. This poses a problem for the nomadic user, who may want to access the messages from many different machines (office PC, home PC, etc.). In the download and keep configuration, messages are not deleted after the user retrieves the messages. This can also be inconvenient, as each time the user retrieves the stored messages from a new machine, all of non-deleted messages will be transferred to the new machine (including very old messages).
19. Yes an organization’s mail server and Web server can have the same alias for a host name. The MX record is used to map the mail server’s host name to its IP address.
20. It is not necessary that Bob will also provide chunks to Alice. Alice has to be in the top 4 neighbors of Bob for Bob to send out chunks to her; this might not occur even if Alice is provides chunks to Bob throughout a 30-second interval.
21. Alice will get her first chunk as a result of she being selected by one of her neighbors as a result of an “optimistic unchoke,” for sending out chunks to her.

22. The overlay network in a P2P file sharing system consists of the nodes participating in the file sharing system and the logical links between the nodes. There is a logical link (an “edge” in graph theory terms) from node A to node B if there is a semi-permanent TCP connection between A and B. An overlay network does not include routers. With Gnutella, when a node wants to join the Gnutella network, it first discovers (“out of band”) the IP address of one or more nodes already in the network. It then sends join messages to these nodes. When the node receives confirmations, it becomes a member of the of Gnutella network. Nodes maintain their logical links with periodic refresh messages.
23. It is a hybrid of client server and P2P architectures:
- a) There is a centralized component (the index) like in the case of a client server system.
  - b) Other functions (except the indexing) do not use any kind of central server. This is similar to what exists in a P2P system.
24. Mesh DHT: The advantage is to route a message to the peer closest to the key, only one hop is required; the disadvantage is that each peer must track all other peers in the in the DHT. Circular DHT: the advantage is that each peer needs to track only a few other peers; the disadvantage is that  $O(N)$  hops are needed to route a message to a peer responsible for the key.
25. a) User location  
b) NAT traversal
26. a) File Distribution  
b) Instant Messaging  
c) Video Streaming  
d) Distributed Computing
27. With the UDP server, there is no welcoming socket, and all data from different clients enters the server through this one socket. With the TCP server, there is a welcoming socket, and each time a client initiates a connection to the server, a new socket is created. Thus, to support  $n$  simultaneous connections, the server would need  $n+1$  sockets.
28. For the TCP application, as soon as the client is executed, it attempts to initiate a TCP connection with the server. If the TCP server is not running, then the client will fail to make a connection. For the UDP application, the client does not initiate connections (or attempt to communicate with the UDP server) immediately upon execution

## Chapter 2 Problems

### Problem 1

- a) F
- b) T
- c) F
- d) F
- e) F

### Problem 2

Access control commands:

**USER, PASS, ACT, CWD, CDUP, SMNT, REIN, QUIT.**

Transfer parameter commands:

**PORT, PASV, TYPE STRU, MODE.**

Service commands:

**RETR, STOR, STOU, APPE, ALLO, REST, RNFR, RNT0, ABOR, DELE, RMD, MRD, PWD, LIST, NLST, SITE, SYST, STAT, HELP, NOOP.**

### Problem 3

Application layer protocols: DNS and HTTP

Transport layer protocols: UDP for DNS; TCP for HTTP

### Problem 4

- a) The document request was `http://gaia.cs.umass.edu/cs453/index.html`. The Host : field indicates the server's name and `/cs453/index.html` indicates the file name.
- b) The browser is running HTTP version 1.1, as indicated just before the first `<cr><lf>` pair.
- c) The browser is requesting a persistent connection, as indicated by the Connection: keep-alive.
- d) This is a trick question. This information is not contained in an HTTP message anywhere. So there is no way to tell this from looking at the exchange of HTTP messages alone. One would need information from the IP datagrams (that carried the TCP segment that carried the HTTP GET request) to answer this question.

e) Mozilla/5.0. The browser type information is needed by the server to send different versions of the same object to different types of browsers.

### Problem 5

- a) The status code of 200 and the phrase OK indicate that the server was able to locate the document successfully. The reply was provided on Tuesday, 07 Mar 2008 12:39:45 Greenwich Mean Time.
- b) The document index.html was last modified on Saturday 10 Dec 2005 18:27:46 GMT.
- c) There are 3874 bytes in the document being returned.
- d) The first five bytes of the returned document are : <!doc. The server agreed to a persistent connection, as indicated by the Connection: Keep-Alive field

### Problem 6

- a) Persistent connections are discussed in section 8 of RFC 2616 (the real goal of this question was to get you to retrieve and read an RFC). Sections 8.1.2 and 8.1.2.1 of the RFC indicate that either the client or the server can indicate to the other that it is going to close the persistent connection. It does so by including the connection-token "close" in the Connection-header field of the http request/reply.
- b) HTTP does not provide any encryption services.
- c) (From RFC 2616) "Clients that use persistent connections should limit the number of simultaneous connections that they maintain to a given server. A single-user client SHOULD NOT maintain more than 2 connections with any server or proxy."
- d) Yes. (From RFC 2616) "A client might have started to send a new request at the same time that the server has decided to close the "idle" connection. From the server's point of view, the connection is being closed while it was idle, but from the client's point of view, a request is in progress."

### Problem 7

The total amount of time to get the IP address is

$$RTT_1 + RTT_2 + \dots + RTT_n .$$

Once the IP address is known,  $RTT_o$  elapses to set up the TCP connection and another  $RTT_o$  elapses to request and receive the small object. The total response time is

$$2RTT_o + RTT_1 + RTT_2 + \dots + RTT_n$$

## Problem 8

a)

$$RTT_1 + \dots + RTT_n + 2RTT_o + 8 \cdot 2RTT_o \\ = 18RTT_o + RTT_1 + \dots + RTT_n.$$

b)

$$RTT_1 + \dots + RTT_n + 2RTT_o + 2 \cdot 2RTT_o \\ = 6RTT_o + RTT_1 + \dots + RTT_n$$

c)

$$RTT_1 + \dots + RTT_n + 2RTT_o + RTT_o \\ = 3RTT_o + RTT_1 + \dots + RTT_n.$$

## Problem 9

- a) The time to transmit an object of size  $L$  over a link of rate  $R$  is  $L/R$ . The average time is the average size of the object divided by  $R$ :

$$A = (850,000 \text{ bits}) / (15,000,000 \text{ bits/sec}) = .0567 \text{ sec}$$

The traffic intensity on the link is given by  $\beta A = (16 \text{ requests/sec})(.0567 \text{ sec/request}) = 0.907$ . Thus, the average access delay is  $(.0567 \text{ sec}) / (1 - .907) \approx .6 \text{ seconds}$ . The total average response time is therefore  $.6 \text{ sec} + 3 \text{ sec} = 3.6 \text{ sec}$ .

- b) The traffic intensity on the access link is reduced by 60% since the 60% of the requests are satisfied within the institutional network. Thus the average access delay is  $(.0567 \text{ sec}) / [1 - (.4)(.907)] = .089 \text{ seconds}$ . The response time is approximately zero if the request is satisfied by the cache (which happens with probability .6); the average response time is  $.089 \text{ sec} + 3 \text{ sec} = 3.089 \text{ sec}$  for cache misses (which happens 40% of the time). So the average response time is  $(.6)(0 \text{ sec}) + (.4)(3.089 \text{ sec}) = 1.24 \text{ seconds}$ . Thus the average response time is reduced from 3.6 sec to 1.24 sec.

## Problem 10

Note that each downloaded object can be completely put into one data packet. Let  $T_p$  denote the one-way propagation delay between the client and the server.

First consider parallel downloads via non-persistent connections. Parallel download would allow 10 connections share the 150 bits/sec bandwidth, thus each gets just 15 bits/sec. Thus, the total time needed to receive all objects is given by:

$$(200/15 + T_p + 200/15 + T_p + 200/15 + T_p + 100,000/15 + T_p) \\ + (200/(150/10) + T_p + 200/(150/10) + T_p + 200/(150/10) + T_p + 100,000/(150/10) + T_p) \\ = 7377 + 8 \cdot T_p \text{ (seconds)}$$



Then consider persistent HTTP connection. The total time needed is give by:

$$(200/150+T_p + 200/150 +T_p + 200/150+T_p + 100,000/150+ T_p ) \\ + 10*(200/150+T_p + 100,000/150+ T_p ) \\ =7351 + 24*T_p \text{ (seconds)}$$

Assume the speed of light is  $300*10^6$  m/sec, then  $T_p=10/(300*10^6)=0.03$  microsec.  $T_p$  is negligible compared with transmission delay.

Thus, we see that the persistent HTTP does not have significant gain (less than 1 percent) over the non-persistent case with parallel download.

### Problem 11

- a). Yes, because Bob has more connections, so he can proportionally get more aggregate bandwidth share out of the total link bandwidth.
- b) Yes, Bob still needs to perform parallel download, otherwise he will get less bandwidth share than other four users. In fact, all users might tend to open more connections in order to gain more bandwidth share.

### Problem 12

#### TCPServer.java

```
import java.io.*;
import java.net.*;

class TCPServer {
    public static void main(String argv[]) throws Exception
    {
        String clientSentence;
        ServerSocket welcomeSocket = new ServerSocket(6789);
        while(true) {
            Socket connectionSocket = welcomeSocket.accept();

            BufferedReader inFromClient = new BufferedReader(new
            InputStreamReader(connectionSocket.getInputStream( ) ) );

            clientSentence = inFromClient.readLine();

            System.out.println("RECEIVED FROM CLIENT : " +
            clientSentence + "\n");
        }
    }
}
```

### Problem 13

The MAIL FROM: in SMTP is a message from the SMTP client that identifies the sender of the mail message to the SMTP server. The From: on the mail message itself is NOT an SMTP message, but rather is just a line in the body of the mail message.

### Problem 14

SMTP uses a line containing only a period to mark the end of a message body. HTTP uses "Content-Length header field" to indicate the length of a message body. No, HTTP cannot use the method used by SMTP, because HTTP message could be binary data, whereas in SMTP, the message body must be in 7-bit ASCII format.

### Problem 15

MTA stands for Mail Transfer Agents. A mail is forwarded by a source to a MTA and then it follows a sequence of MTAs to reach the receiver's mail reader. We see that this spam email follows a chain of MTAs. An honest MTA should report where it receives the message. Notice that in this email, "asusus-4b96 ([58.88.21.177])" does not report where it receives the email. As we assume that the only the originator is dishonest, so "asusus-4b96 ([58.88.21.177])" must be the originator.

### Problem 16

UIDL abbreviates "unique-ID listing". When a POP3 client issues the UIDL command, the server responds with the unique message ID for all of the messages present in the users mailbox. This command is useful for "download and keep". By keeping a file that lists the messages retrieved in earlier sessions, the client can use the UIDL command to determine which messages on the server have already been seen.

### Problem 17

a) C: dele 1  
C: retr 2  
S: (blah blah ...  
S: .....blah)  
S: .  
C: dele 2  
C: quit  
S: +OK POP3 server signing off

- b) C: retr 2  
 S: blah blah ...  
 S: .....blah  
 S: .  
 C: quit  
 S: +OK POP3 server signing off
- c) C: list  
 S: 1 498  
 S: 2 912  
 S: .  
 C: retr 1  
 S: blah .....  
 S: ....blah  
 S: .  
 C: retr 2  
 S: blah blah ...  
 S: .....blah  
 S: .  
 C: quit  
 S: +OK POP3 server signing off

## Problem 18

- a) For a given input of domain name (such as ccn.com), IP address or network administrator name, *whois* database can be used to locate the corresponding registrar, whois server, DNS server, and so on.
- b) NS4.YAHOO.COM from www.register.com; NS1.MSFT.NET from ww.register.com
- c) *Local Domain: www.mindspring.com*  
 Web servers : www.mindspring.com  
                   207.69.189.21, 207.69.189.22,  
                   207.69.189.23, 207.69.189.24,  
                   207.69.189.25, 207.69.189.26, 207.69.189.27,  
                   207.69.189.28  
 Mail Servers : mx1.mindspring.com (207.69.189.217)  
                   mx2.mindspring.com (207.69.189.218)  
                   mx3.mindspring.com (207.69.189.219)  
                   mx4.mindspring.com (207.69.189.220)  
 Name Servers: itchy.earthlink.net (207.69.188.196)  
                   scratchy.earthlink.net (207.69.188.197)

*www.yahoo.com*

Web Servers: www.yahoo.com (216.109.112.135, 66.94.234.13)

Mail Servers: a.mx.mail.yahoo.com (209.191.118.103)  
b.mx.mail.yahoo.com (66.196.97.250)  
c.mx.mail.yahoo.com (68.142.237.182, 216.39.53.3)  
d.mx.mail.yahoo.com (216.39.53.2)  
e.mx.mail.yahoo.com (216.39.53.1)  
f.mx.mail.yahoo.com (209.191.88.247, 68.142.202.247)  
g.mx.mail.yahoo.com (209.191.88.239, 206.190.53.191)

Name Servers: ns1.yahoo.com (66.218.71.63)  
ns2.yahoo.com (68.142.255.16)  
ns3.yahoo.com (217.12.4.104)  
ns4.yahoo.com (68.142.196.63)  
ns5.yahoo.com (216.109.116.17)  
ns8.yahoo.com (202.165.104.22)  
ns9.yahoo.com (202.160.176.146)

*www.hotmail.com*

Web Servers: www.hotmail.com (64.4.33.7, 64.4.32.7)

Mail Servers: mx1.hotmail.com (65.54.245.8, 65.54.244.8, 65.54.244.136)  
mx2.hotmail.com (65.54.244.40, 65.54.244.168, 65.54.245.40)  
mx3.hotmail.com (65.54.244.72, 65.54.244.200, 65.54.245.72)  
mx4.hotmail.com (65.54.244.232, 65.54.245.104, 65.54.244.104)

Name Servers: ns1.msft.net (207.68.160.190)  
ns2.msft.net (65.54.240.126)  
ns3.msft.net (213.199.161.77)  
ns4.msft.net (207.46.66.126)  
ns5.msft.net (65.55.238.126)

- d) The yahoo web server has multiple IP addresses  
www.yahoo.com (216.109.112.135, 66.94.234.13)
- e) The address range for Polytechnic University: 128.238.0.0 – 128.238.255.255
- f) An attacker can use the *whois* database and nslookup tool to determine the IP address ranges, DNS server addresses, etc., for the target institution.
- g) By analyzing the source address of attack packets, the victim can use whois to obtain information about domain from which the attack is coming and possibly inform the administrators of the origin domain.

## Problem 19

- a)  
The following delegation chain is used for gaia.cs.umass.edu

a.root-servers.net  
E.GTLD-SERVERS.NET  
ns1.umass.edu(authoritative)

First command:

dig +norecurse @a.root-servers.net any gaia.cs.umass.edu

;; AUTHORITY SECTION:

edu.	172800	IN	NS	E.GTLD-SERVERS.NET.
edu.	172800	IN	NS	A.GTLD-SERVERS.NET.
edu.	172800	IN	NS	G3.NSTLD.COM.
edu.	172800	IN	NS	D.GTLD-SERVERS.NET.
edu.	172800	IN	NS	H3.NSTLD.COM.
edu.	172800	IN	NS	L3.NSTLD.COM.
edu.	172800	IN	NS	M3.NSTLD.COM.
edu.	172800	IN	NS	C.GTLD-SERVERS.NET.

Among all returned edu DNS servers, we send a query to the first one.

dig +norecurse @E.GTLD-SERVERS.NET any gaia.cs.umass.edu

umass.edu.	172800	IN	NS	ns1.umass.edu.
umass.edu.	172800	IN	NS	ns2.umass.edu.
umass.edu.	172800	IN	NS	ns3.umass.edu.

Among all three returned authoritative DNS servers, we send a query to the first one.

dig +norecurse @ns1.umass.edu any gaia.cs.umass.edu

gaia.cs.umass.edu. 21600 IN A 128.119.245.12

b) The answer for google.com could be:

a.root-servers.net  
E.GTLD-SERVERS.NET  
ns1.google.com(authoritative)

## Problem 20

We can periodically take a snapshot of the DNS caches in those local DNS servers. The Web server that appears most frequently in the DNS caches is the most popular server. This is because if more users are interested in a Web server, then DNS requests for that server are more frequently sent by users. Thus, that Web server will appear in the DNS caches more frequently.

For a complete measurement study, see:

Craig E. Wills, Mikhail Mikhailov, Hao Shang

“Inferring Relative Popularity of Internet Applications by Actively Querying DNS Caches”, in IMC'03, October 27-29, 2003, Miami Beach, Florida, USA

## Problem 21

Yes, we can use dig to query that Web site in the local DNS server.

For example, “dig cnn.com” will return the query time for finding cnn.com. If cnn.com is just accessed a couple of seconds ago, an entry for cnn.com is cached in the local DNS cache, so the query time is 0 msec. Otherwise, the query time is large.

## Problem 22

For calculating the minimum distribution time for client-server distribution, we use the following formula:

$$D_{cs} = \max \{NF/u_s, F/d_{\min}\}$$

Similarly, for calculating the minimum distribution time for P2P distribution, we use the following formula:

$$D_{p2p} = \max \{F/u_s, F/d_{\min}, NF/(u_s + \sum_{i=1}^N u_i)\}$$

Where,  $F = 15 \text{ Gbits} = 15 * 1024 \text{ Mbits}$

$u_s = 30 \text{ Mbps}$

$d_{\min} = d_i = 2 \text{ Mbps}$

**Note, 300Kbps = 300/1024 Mbps.**

### Client Server

		N		
		10	100	1000
u	300 Kbps	7680	51200	512000
	700 Kbps	7680	51200	512000
	2 Mbps	7680	51200	512000

### Peer to Peer

		N		
		10	100	1000
u	300 Kbps	7680	25904	47559
	700 Kbps	7680	15616	21525
	2 Mbps	7680	7680	7680

## Problem 23

a) Consider a distribution scheme in which the server sends the file to each client, in parallel, at a rate of a rate of  $u_s/N$ . Note that this rate is less than each of the client's download rate, since by assumption  $u_s/N \leq d_{\min}$ . Thus each client can also receive at rate

$u_s/N$ . Since each client receives at rate  $u_s/N$ , the time for each client to receive the entire file is  $F/(u_s/N) = NF/u_s$ . Since all the clients receive the file in  $NF/u_s$ , the overall distribution time is also  $NF/u_s$ .

b) Consider a distribution scheme in which the server sends the file to each client, in parallel, at a rate of  $d_{\min}$ . Note that the aggregate rate,  $N d_{\min}$ , is less than the server's link rate  $u_s$ , since by assumption  $u_s/N \geq d_{\min}$ . Since each client receives at rate  $d_{\min}$ , the time for each client to receive the entire file is  $F/d_{\min}$ . Since all the clients receive the file in this time, the overall distribution time is also  $F/d_{\min}$ .

c) From Section 2.6 we know that

$$D_{CS} \geq \max \{NF/u_s, F/d_{\min}\} \quad (\text{Equation 1})$$

Suppose that  $u_s/N \leq d_{\min}$ . Then from Equation 1 we have  $D_{CS} \geq NF/u_s$ . But from (a) we have  $D_{CS} \leq NF/u_s$ . Combining these two gives:

$$D_{CS} = NF/u_s \text{ when } u_s/N \leq d_{\min}. \quad (\text{Equation 2})$$

We can similarly show that:

$$D_{CS} = F/d_{\min} \text{ when } u_s/N \geq d_{\min} \quad (\text{Equation 3}).$$

Combining Equation 2 and Equation 3 gives the desired result.

## Problem 24

a) Define  $u = u_1 + u_2 + \dots + u_N$ . By assumption

$$u_s \leq (u_s + u)/N \quad \text{Equation 1}$$

Divide the file into  $N$  parts, with the  $i^{\text{th}}$  part having size  $(u_i/u)F$ . The server transmits the  $i^{\text{th}}$  part to peer  $i$  at rate  $r_i = (u_i/u)u_s$ . Note that  $r_1 + r_2 + \dots + r_N = u_s$ , so that the aggregate server rate does not exceed the link rate of the server. Also have each peer  $i$  forward the bits it receives to each of the  $N-1$  peers at rate  $r_i$ . The aggregate forwarding rate by peer  $i$  is  $(N-1)r_i$ . We have

$$(N-1)r_i = (N-1)(u_s u_i)/u \leq u_i,$$

where the last inequality follows from Equation 1. Thus the aggregate forwarding rate of peer  $i$  is less than its link rate  $u_i$ .

In this distribution scheme, peer  $i$  receives bits at an aggregate rate of

$$r_i + \sum_{j < i} r_j = u_s$$

Thus each peer receives the file in  $F/u_s$ .

b) Again define  $u = u_1 + u_2 + \dots + u_N$ . By assumption

$$u_s \geq (u_s + u)/N \quad \text{Equation 2}$$

Let  $r_i = u_i/(N-1)$  and  
 $r_{N+1} = (u_s - u/(N-1))/N$

In this distribution scheme, the file is broken into  $N+1$  parts. The server sends bits from the  $i^{\text{th}}$  part to the  $i^{\text{th}}$  peer ( $i = 1, \dots, N$ ) at rate  $r_i$ . Each peer  $i$  forwards the bits arriving at rate  $r_i$  to each of the other  $N-1$  peers. Additionally, the server sends bits from the  $(N+1)^{\text{st}}$  part at rate  $r_{N+1}$  to each of the  $N$  peers. The peers do not forward the bits from the  $(N+1)^{\text{st}}$  part.

The aggregate send rate of the server is

$$r_1 + \dots + r_N + N r_{N+1} = u/(N-1) + u_s - u/(N-1) = u_s$$

Thus, the server's send rate does not exceed its link rate. The aggregate send rate of peer  $i$  is

$$(N-1)r_i = u_i$$

Thus, each peer's send rate does not exceed its link rate.

In this distribution scheme, peer  $i$  receives bits at an aggregate rate of

$$r_i + r_{N+1} + \sum_{j \neq i} r_j = u/(N-1) + (u_s - u/(N-1))/N = (u_s + u)/N$$

Thus each peer receives the file in  $NF/(u_s+u)$ .

(For simplicity, we neglected to specify the size of the file part for  $i = 1, \dots, N+1$ . We now provide that here. Let  $\Delta = (u_s+u)/N$  be the distribution time. For  $i = 1, \dots, N$ , the  $i^{\text{th}}$  file part is  $F_i = r_i \Delta$  bits. The  $(N+1)^{\text{st}}$  file part is  $F_{N+1} = r_{N+1} \Delta$  bits. It is straightforward to show that  $F_1 + \dots + F_{N+1} = F$ .)

c) The solution to this part is similar to that of 17 (c). We know from section 2.6 that

$$D_{p2p} \geq \max\{F/u_s, NF/(u_s + u)\}$$

Combining this with (a) and (b) gives the desired result.



### Problem 25

There are  $N$  nodes in the overlay network. There are  $N(N-1)/2$  edges.

### Problem 26

Yes. His first claim is possible, as long as there are enough peers staying in the swarm for a long enough time. Bob can always receive data through optimistic unchoking by other peers.

His second claim is also true. He can run a client on each machine, and let each client do “free-riding”, and combine those collected chunks from different machines into a single file. He can even write a small scheduling program to let different machines only asking for different chunks of the file. This is actually a kind of Sybil attack in P2P networks.

### Problem 27

a).

Note that we assume  $n_b \geq n_a$ .

$\frac{C(N - n_a, n_b - n_a)}{C(N, n_b)}$ , where  $C(N, n)$  is the notation for combination, which means the number of ways of choosing  $n$  out of  $N$ .

$$b). p(n_a) = \sum_{n_b=n_a}^{N-1} \frac{1}{N} \frac{C(N - n_a, n_b - n_a)}{C(N, n_b)}.$$

$$c). \text{prob} = 1 - \left( \sum_{n_a=0}^{N-1} \frac{1}{N} p(n_a) \right)^5.$$

For a complete analysis, see:

Donyu Qiu and R. Srikant.

Modeling and Performance Analysis of BitTorrent-Like Peer-to-Peer Networks.

ACM Sigcomm 2004, Portland, Oregon, USA

### Problem 28

Peer 3 learns that peer 5 has just left the system, so Peer 3 asks its first successor (peer 4) for the identifier of its immediate successor (peer 8). Then peer 3 will make peer 8 as its second successor.

Note that peer 3 knows that peer 5 was originally the first successor of peer 4, so peer 3 would wait until peer 4 finishes updating its first successor.

## Problem 29

Peer 6 would first send peer 15 a message, saying “what will be peer 6’s predecessor and successor?” This message gets forwarded through the DHT until it reaches peer 5, who realizes that it will be 6’s predecessor and that its current successor, peer 8, will become 6’s successor. Next, peer 5 sends this predecessor and successor information back to 6. Peer 6 can now join the DHT by making peer 8 its successor and by notifying peer 5 that it should change its immediate successor to 6.

## Problem 30

a).

Our assumption about keys and queries:

1. All keys are uniformly at random distributed in the key range, and all 8 peers are responsible for the same number of queries on average.
2. The queries generated by a peer are for keys uniformly at random distributed in the key range. That is, the query for any key is generated with the same probability.

Because of the homogeneity of peers and queries, we know that all peers will choose a shortcut peer with the same number of *overlay hops* away.

We also assume that *by default*, a peer node *only* knows about its immediate successor peer and its immediate predecessor node. (Unlike the case where a peer must know its second immediate successor in order to deal with peer churn).

We further assume that a peer can forward query to its predecessor.

And if there are multiple routing paths exist for a query, a peer always chooses the shortest path.

Note the number of messages sent for a query for a key is equal to the number of *routing hops* needed from the query generating peer to the peer that is holding the key.

Note that in our description, a routing hop is different from an overlay hop. An overlay hop simply means a logical hop between two *adjacent* peers along the DHT overlay ring. But a routing hop can span multiple overlay hops (or multiple consecutive adjacent peers) if shortcut is allowed.

Thus, minimizing the number of messages sent for any query (starting from any peer) is equivalent to minimizing the average or total number of routing hops traversed from one peer to all other peers.

Without loss of generality, let's look at peer 0.

To solve this problem, we look at all possibilities: a node shortcuts to a peer two overlay hops away in DHT id ring; or three overlay hops away; or four overlay hops away, so on. We can find that *the best configuration is for a peer to choose a shortcut peer with 4 overlay hops away*. "Best" in the sense that the average number of messages per query (or the total number of routing hops for all queries for all keys in the key space) is minimized. The computation is shown in the following table. Each column (except the last column) shows the number of messages needed for routing a query within a range. We see that the total number of messages needed is 11 when every peer shortcuts to another peer with 4 overlay hops (i.e., span 4 consecutive adjacent peers) away.

Queries for keys in range		(0,8]	(8,16]	(16,24]	(24,32]	(32,40]	(40,48]	(48,56]	(56,0]	Total messages
Number of overlay hops away for shortcutting	2	1	1	2	2	3	3	1	0	13
	3	1	2	1	2	3	2	1	0	12
	4	1	2	2	1	2	2	1	0	11
	5	Same as 3 hops away								12
	6	Same as 2 hops away								13
<p>Note, if there are multiple routing paths exist for a query, we choose the shortest path's length as the number of needed messages.</p> <p>So for example, for the case where a peer shortcuts to another peer 4 overlay hops away, and if peer 0's query is for a key in range (48,56], then the shortest path is 0→56, only one routing hop (i.e., one message).</p>										

b).

Follow the same reasoning as in part a).

We can find that *the best configuration is for a peer to choose two shortcut peers with 3 and 6 overlay hops away, or two shortcut peers with 5 and 6 overlay hops away.*

Queries for keys in range		(0,8]	(8,16]	(16,24]	(24,32]	(32,40]	(40,48]	(48,56]	(56,0]	Total messages
Num. of overlay hops away										
Nbr 1	Nbr 2									
2	3	1	1	1	2	2	2	1	0	10
2	4	1	1	2	2	2	1	1	0	10
2	5	1	1	1	2	2	2	1	0	10
2	6	1	1	1	2	2	2	1	0	10
3	4	1	2	1	1	1	2	1	0	9
3	5	1	2	1	1	1	2	1	0	9
3	6	1	1	1	2	1	1	1	0	<b>8</b>
4	5	Same as nbr 1, nbr 2 with 3 and 4 hops away respectively								9
4	6	1	1	2	1	2	1	1	0	9
5	6	Same as nbr 1, nbr 2 with 3 and 6 hops away respectively								<b>8</b>

## Problem 31

For each key, we first calculate the distances (according to  $d(k,p)$ ) between itself and all peers, and then store this key into the peer that is closed to this key (with smallest distance value).

Then, as in circular DHT described in textbook, we arrange those peers in a ring. In this ring, there are many clusters of keys, and each cluster is centered at a particular peer. Each key in a peer's cluster is closer to this peer than to all other peers. Some of the keys in this cluster can be larger than this peer's ID. Note that a peer is responsible for the keys in its cluster, instead of being responsible for only keys that are preceding it (i.e, keys have smaller value than the ID of this peer) in the key range.

Each peer keeps a routing table of  $n$  lists of entries. Each entry contains the information of one other peer. These lists are arranged sequentially, and the  $k$ -th ( $1 \leq k \leq n$ ) list contains peers with IDs that differ from that of this peer in  $k$ -th significant bit but match with all  $k-1$  bits more significant than  $k$ , including the most significant bit, the second most significant bit, so on, until the  $(k-1)$ -th most significant bit. Note here we use longest-prefix matching. Also note that with this arrangement, it is possible that half of the IDs in the ID range can be put into the first list.

If  $i > j$ , then a peer in  $i$ -th list is closer to this node than a peer in  $j$ -th list.

The query routing can be done as follows. A peer first tries to match the bits of the key with its own ID's bits, and finds the "right" list in its routing table, and then forwards the query to any one entry in the list. A "right" list is a list that has the longest prefix matching with the target key. Once a peer receives the target key, it also checks its routing table, and forwards the search query to a peer in the "right" list, so on, so forth, until a peer that is responsible for the key is located, or returns "no such key" if no further routing is possible.

### Problem 32

This is a generalized Kademlia DHT, and also similar to Pastry's prefix-matching DHT. The DHT based on binary numbers generates more messages,  $\log_2 N$  in the worst case. This new DHT generates  $\log_b N$  messages in the worst case.

### Problem 33

Yes, that assignment scheme of keys to peers does not consider underlying network at all, so it very likely causes mismatch.

The mismatch may potentially degrade the search performance. For example, consider a logical path  $p1$  (consisting of only two logical links):  $A \rightarrow B \rightarrow C$ , where  $A$  and  $B$  are neighboring peers, and  $B$  and  $C$  are neighboring peers. Suppose that there is another logical path  $p2$  from  $A$  to  $B$  (consisting of 3 logical links):  $A \rightarrow D \rightarrow E \rightarrow C$ .

It might be the case that  $A$  and  $B$  could be very far away physically, and  $B$  and  $C$  could be very far away physically. But  $A$ ,  $D$ ,  $E$ , and  $C$  are very close physically. In other words, a shorter logical path corresponds to a longer physical path than does a longer logical path.

### Problem 34

a) If you run TCPClient first, then the client will attempt to make a TCP connection with a non-existent server process. A TCP connection will not be made.

b) UDPClient doesn't establish a TCP connection with the server. Thus, everything should work fine if you first run UDPClient, then run UDPServer, and then type some input into the keyboard.

c) If you use different port numbers, then the client will attempt to establish a TCP connection with the wrong process or a non-existent process. Errors will occur.

### **Problem 35**

With the original line, UDPClient does not specify a port number when it creates the socket. In this case, the code lets the underlying operating system choose a port number. With the replacement line, when UDPClient is executed, a UDP socket is created with port number 5432 .

UDPServer needs to know the client port number so that it can send packets back to the correct client socket. Glancing at UDPServer, we see that the client port number is not “hard-wired” into the server code; instead, UDPServer determines the client port number by unraveling the datagram it receives from the client (using the .getPort() method). Thus UDP server will work with any client port number, including 5432. UDPServer therefore does not need to be modified.

Before:

Client socket = x (chosen by OS)  
Server socket = 9876

After:

Client socket = 5432