3. Introduction to the Transport Layer and Socket Programming

Basic concepts

- The UDP transport protocol
- Building a UDP client and server in Java
- The TCP transport protocol
- Building a TCP client and server in Java

From Jon Turner – based partly on material from Kurose & Ross



- Routers forward packets among hosts
- Internet Protocol (IP) provides best-effort delivery of datagram packets based on addresses in packet headers
- Processes use sockets as interface to network stack
- Transport protocols define port numbers used to identify specific sockets

Transport Services and Protocols

- Enable communication between application processes running on different hosts
 - » most basic function is extending addressing beyond hosts
 - » in Internet, port numbers are used for this purpose
- Transport protocols run in end systems
 - » although in Internet, some network components (e.g. NATs) do examine and even modify transport fields
- Two principal transport protocols in Internet
 - » UDP provides datagram interface, port numbers and some limited error detection
 - » TCP provides byte-stream interface, port numbers, reliable data transport, flow control, congestion control
 - » neither provides bandwidth or delay guarantees
 - since the Internet (typically) does not

Transport Layer (De)Multiplexing

Multiplexing, i.e., combining over shared resource

- » Transport connections share the same IP address
- » need mechanism to identify individual programs port numbers
- » running programs communicate through sockets and transport protocol software maps port numbers to sockets

Connectionless (de)multiplexing

- » to send packet to a remote program, must know the port number it is using, in addition to its host's IP address
- » when a UDP packet arrives at a host, the *destination port* # is used to identify the socket that is to receive the packet
 - the destination port is the only information used to identify socket
 - so, packets from different remote hosts can be delivered to the same socket
- » the remote program uses the source IP address and port # of received packet when sending a reply back to the sender

UDP Details

Port number are 16 bits long	16 bits	16 bits		
	src port	dst port		
Length specifies number of bytes in UDP "segment", including header	length	checksum		
Checksum is used to detect errors in				
UDP segment				
 » sender computes checksum by adding 16 bit chunks (keep adding beyond 16 bits) • On overflow, add carry bits back to 16 bits value • Take one's complement (change 0/1 to 1/0) 				
» receiver checks for errors by re-computing sum and comparing to value in packet				
» since IP runs over multiple different link layer protocols, it cannot rely on link layer for error detection				
 even though link layer routinely implements error 	or detection	on 5		

One's Complement Arithmetic

- A: 0000 1000 1000 1001
 - B: 1000 0000 0001 0001
 - C: 1001 1111 0001 1111
- A+B= 1000 1000 1001 1010
 - +C: 1001 1111 0001 1111
 - **1**001**0 0**111 101**1 100**1
 - 0001
 - = 0010 0111 1011 1010
 - = 1101 1000 0100 0101 (one's complement)

Exercises

- 1. What services does TCP provide that UDP does not?
- 2. Name three services that neither UDP nor TCP provide.
- 3. Suppose a host is running two programs A and B that both communicate over the internet using UDP. If the socket for program A is mapped to port number 6789, is it possible for the socket for program B to also be mapped to port number 6789?
- 4. When a UDP server receives a packet from a client program and replies to the client, how does it know where to send the packet?
- 5. Assume a UDP payload that consists of the 6 bytes on the previous slide. Can you change one bit without changing the checksum? What about 2 bits? Can you generalize this result?

Simple UDP Echo Server in Java

```
import java.io.*; import java.net.*;
public class UdpServer {
   public static void main(String args[]) throws Exception {
      // open datagram socket on port 9876
      DatagramSocket sock = new DatagramSocket(9876);
      // create two packets sharing a common buffer
      byte[] buf = new byte[1000];
      DatagramPacket inPkt = new DatagramPacket(buf, buf.length);
      DatagramPacket outPkt = new DatagramPacket(buf, buf.length);
      while (true) {
          // wait for incoming packet
          sock.receive(inPkt);
          // set address, port and length fields of outPkt
          // so as to return contents of inPkt to sender
          outPkt.setAddress(inPkt.getAddress());
          outPkt.setPort(inPkt.getPort());
          outPkt.setLength(inPkt.getLength());
          // and send it back
          sock.send(outPkt);
}
      }
   }
```

UDP Echo Client

```
import java.io.*; import java.net.*;
public class UdpClient {
    public static void main(String args[]) throws Exception {
        // get server address and open socket
        InetAddress serverAdr = InetAddress.getByName(args[0]);
```

DatagramSocket sock = new DatagramSocket();

```
// build packet addressed to server, then send it
```

byte[] outBuf = args[1].getBytes("US-ASCII");

DatagramPacket outPkt = new

DatagramPacket(outBuf,outBuf.length,serverAdr,9876); sock.send(outPkt);

```
// create buffer and packet for reply, then receive it
byte[] inBuf = new byte[1000];
```

DatagramPacket inPkt = new DatagramPacket(inBuf,inBuf.length); sock.receive(inPkt);

// print buffer contents and close socket

String reply = new String(inBuf,0,inPkt.getLength(),"US-ASCII");
System.out.println(reply);
sock.close();

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Recap of Classes/Methods Used

DatagramSocket (defined in java.net)

- » used to read/write packets
- » opened by constructor, closed using close() method

DatagramPacket (in java.net)

- » defines packet with payload defined by array of bytes
 - in Java, a byte is not the same thing as a char
 - encode strings as bytes using String's getBytes method with a specified encoding
 - convert byte arrays to Strings using String constructor with specified encoding
- » address and port fields specify peer address/port
- » length field is # of bytes to send in packet, or # received
- InetAddress (in java.net)
 - » use static method getByName(addressString) to convert a destination host's name to its internet address

Exercises

- In UdpEchoServer, two packet objects are used. Do you think this is necessary? If so, explain why. If not, show how you can modify the code to use just one packet object.
- UdpEchoClient uses the US-ASCII method to encode Strings as bytes. If you run it using the command

java UdpEchoClient serverHostName "hi there"

how many bytes are carried in the payload of the UDP packet? Suppose you modified the program to use UTF-16, in place of US-ASCII, how many bytes would be carried in the UDP payload in this case? Try it.

Key TCP Concepts

- To provide reliable communication, the TCP networking software at both communicating hosts must cooperate
 - » requires information that must be maintained consistently at both ends
 - » to allow the network software to initialize this information, communicating processes must first "establish a connection"

Two kinds of sockets in TCP

- » listening socket used by server to make itself available for connection requests
 - any host in the internet can send a connection request to a listening socket (assuming it knows the IP address and port#)

» connection socket used by a client-server pair to exchange data

- identified by "4-tuple": local address/port#, remote address/port#
- server can have multiple connection sockets using same port#
- typically, each connection socket handled by separate thread

TCP Connection Establishment



open socket and configure as "listening" socket accept (and wait)

connection socket opened source address, port associated with socket

now connected

}

TCP Echo Server

```
import java.io.*; import java.net.*;
public class TcpServer {
 public static void main(String args[]) throws Exception {
      // create listen socket and buffer
      ServerSocket listenSock = new ServerSocket(6789);
      byte[] buf = new byte[1000];
      while (true) {
              // wait for connection request; create connection socket
              Socket connSock = listenSock.accept();
              // create buffered versions of socket's in/out streams
              BufferedInputStream in = new BufferedInputStream(
                                           connSock.getInputStream());
              BufferedOutputStream out = new BufferedOutputStream(
                                           connSock.getOutputStream());
              while (true) { // read and echo until peer closes
                     int nbytes = in.read(buf, 0, buf.length);
                     if (nbytes < 0) break;
                     out.write(buf,0,nbytes); out.flush();
              }
              connSock.close();
```

}

}

TCP Echo Client

```
import java.io.*; import java.net.*;
public class TcpClient {
   public static void main(String args[]) throws Exception {
      // open socket and connect to server
      Socket sock = new Socket(args[0], 6789);
```

// output second argument to server, adding a newline
// as delimiter; flush all data from buffer to network
out.write(args[1]); out.newLine(); out.flush();

// read data up to end of line and close connection
System.out.println(in.readLine());
sock.close();

Additional Classes/Methods Used

ServerSocket (in java.net) » used to listen for incoming connection requests Socket (in java.net) » used for connection sockets » associated input and output streams for transferring bytes BufferedInputStream/BufferedOutputStream (in java.io) » buffers bytes in user space for more efficient data transfers InputStreamReader/OutputStreamWriter (in java.io) » for converting between bytes and chars/Strings » can use variety of encoding methods BufferedReader/BufferedWriter (in java.io) » buffers chars in user space for more efficient transfers » readLine() method simplifies parsing of input 16

Engineering

TCP Provides a Byte Stream Interface



Bytes that are written together may not be transferred together, so don't expect single read to get them all

If receiver is slow to read bytes

- » receiver's socket buffer fills, causing it to flow-control sender
- » when sender's socket buffer fills, additional writes block, suspending application program
- » bytes are transferred as space becomes available at receiver

Formatting a Byte Stream

- Since no visible packet boundaries (as with datagrams), sending program must format data to assist receiver
- For character data, common technique is to use newline character (or CR,LF pair) as delimiter
 - » receiver reads complete lines, then parses contents of each line
- For binary data, simplest approach is fixed format
 - » receiver relies on pre-arranged format when reading data
 - » can send a variable length array by first sending a length field
 - receiver first reads length, then reads as many additional items as specified by the length field
- Many other possibilities, but there must be some welldefined format that receiver can depend on

Note: Network and Host Byte Orders

- Two common conventions for representing multi-byte values in a computer's memory
 - » little-endian low order byte comes first in memory
 - so 32 bit hex value 0xabcdef12 is stored 12, ef, cd, ab
 - » big-endian high order byte comes first

Internet protocols assume high-order byte sent first

- » to ensure this, may need to reformat data before sending a packet (and after receiving it)
- » Java provides mechanisms to handle this automatically
 - DataInputStream/DataOutputStream (in java.io) for TCP sockets declare by "wrapping" BufferedInputStream/OutputStream DataInputStream in = new DataInputStream(new BufferedInputStream(connSock.getInputStream())); use provided methods to write/read binary data int i = in.readInt(); float x; out.writeFloat(x);
 - ByteBuffers (in java.nio) for UDP sockets

TCP Segment Format

- Port numbers are 16 bits long
- Length specifies number of 32 bit words in TCP header (typically 5)
- Flags include
 - » SYN and FIN bits used to setup and teardown connections
- Checksum is used to detect errors in TCP segment
 - » sender computes checksum by adding 16 bit chunks
 - on arithmetic overflow, sum is incremented

Engineering

16 bits			16 bits	
src port		ort	dst port	
sequence #				
acknowledgment #				
hdr leng	-	flags	recv window	
checksum		um	urgent data	
options				
application data				

- » receiver re-computes sum and compares to value in packet
- Defer discussion of other fields

Exercises

- How many sockets does a UDP server need to communicate with 100 remote clients concurrently? How many port numbers does it need?
- 2. How many sockets does a TCP server need to communicate with 100 remote clients concurrently? How many port numbers does it need?
- 3. How does the network stack identify the socket to which an incoming TCP packet is to be delivered?