Chapter 3 Transport Layer

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Computer Networking: A Top Down Approach 4th edition. Jim Kurose, Keith Ross Addison-Wesley, July 2007.

<u>Chapter 3 outline</u>

- 3.1 Transport-layer services
- 3.2 Multiplexing and demultiplexing
- 3.3 Connectionless transport: UDP
- 3.4 Principles of reliable data transfer

- 3.5 Connection-oriented transport: TCP
 - segment structure
 - o reliable data transfer
 - flow control
 - connection management
- 3.6 Principles of congestion control
- 3.7 TCP congestion control

<u>TCP reliable data transfer</u>

- TCP creates rdt service on top of IP's unreliable service
- Pipelined segments
- Cumulative acks
- TCP uses single retransmission timer

- Retransmissions are triggered by:
 - timeout events
 - duplicate acks
- Initially consider simplified TCP sender:
 - ignore duplicate acks
 - ignore flow control, congestion control

TCP sender events:

data rcvd from app:

- Create segment with seq #
- seq # is byte-stream number of first data byte in segment
- start timer if not already running (think of timer as for oldest unacked segment)
- **cxpiration interval:** TimeOutInterval

<u>timeout:</u>

- retransmit segment that caused timeout
- restart timer

Ack rcvd:

- If acknowledges previously unacked segments
 - update what is known to be acked
 - start timer if there are outstanding segments

NextSeqNum = InitialSeqNum SendBase = InitialSeqNum

} /* end of loop forever */

loop (forever) {
 switch(event)

event: data received from application above create TCP segment with sequence number NextSeqNum if (timer currently not running) start timer pass segment to IP NextSeqNum = NextSeqNum + length(data)

```
event: timer timeout
retransmit not-yet-acknowledged segment with
smallest sequence number
start timer
```

```
event: ACK received, with ACK field value of y
if (y > SendBase) {
    SendBase = y
    if (there are currently not-yet-acknowledged segments)
        start timer
    }
}
```

<u>TCP</u> <u>sender</u> (simplified)

Comment: • SendBase-1: last cumulatively ack'ed byte Example: • SendBase-1 = 71; y= 73, so the rcvr wants 73+; y > SendBase, so that new data is acked

Fast Retransmit

- Time-out period often relatively long:
 - long delay before resending lost packet
- Detect lost segments via duplicate ACKs.
 - Sender often sends many segments back-toback
 - If segment is lost, there will likely be many duplicate ACKs.

If sender receives 3 ACKs for the same data, it supposes that segment after ACKed data was lost:

> <u>fast retransmit</u>: resend segment before timer expires

Fast retransmit algorithm:



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TCP Flow Control

receive side of TCP connection has a receive buffer:

flow control-

sender won't overflow receiver's buffer by transmitting too much, too fast



app process may be slow at reading from buffer speed-matching service: matching the send rate to the receiving app's drain rate

TCP Flow control: how it works



- (Suppose TCP receiver discards out-of-order segments)
- □ spare room in buffer
- = RcvWindow
- = RcvBuffer-[LastByteRcvd -LastByteRead]

Rcvr advertises spare room by including value of RcvWindow in segments

- Sender limits unACKed data to RcvWindow
 - guarantees receive buffer doesn't overflow

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TCP Connection Management

- <u>Recall:</u> TCP sender, receiver establish "connection" before exchanging data segments
- initialize TCP variables:
 - o seq. #s
 - buffers, flow control info (e.g. RcvWindow)
- client: connection initiator
 Socket clientSocket = new
 Socket("hostname","port
 number");
- Socket connectionSocket =
 welcomeSocket.accept();

Three way handshake:

- <u>Step 1:</u> client host sends TCP SYN segment to server
 - specifies initial seq #
 - o no data
- <u>Step 2:</u> server host receives SYN, replies with SYNACK segment
 - server allocates buffers
 - specifies server initial seq. #
- <u>Step 3:</u> client receives SYNACK, replies with ACK segment, which may contain data

TCP Connection Management (cont.)

Closing a connection:

client closes socket:
 clientSocket.close();

<u>Step 1:</u> client end system sends TCP FIN control segment to server

<u>Step 2:</u> server receives FIN, replies with ACK. Closes connection, sends FIN.



TCP Connection Management (cont.)

- <u>Step 3:</u> client receives FIN, replies with ACK.
 - Enters "timed wait" will respond with ACK to received FINs
- <u>Step 4:</u> server, receives ACK. Connection closed.
- <u>Note:</u> with small modification, can handle simultaneous FINs.



TCP Connection Management (cont)

