COSC264 Introduction to Computer Networks and the Internet

Transport Layer Protocols: UDP and TCP

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Outline

- Transport-layer services
- Multiplexing and demultiplexing
- Connectionless transport: UDP

Role of Transport Layer

Application layer

- Communication for specific applications
- e.g., HyperText Transfer Protocol (HTTP), File Transfer Protocol (FTP), Network News Transfer Protocol (NNTP)

Transport layer

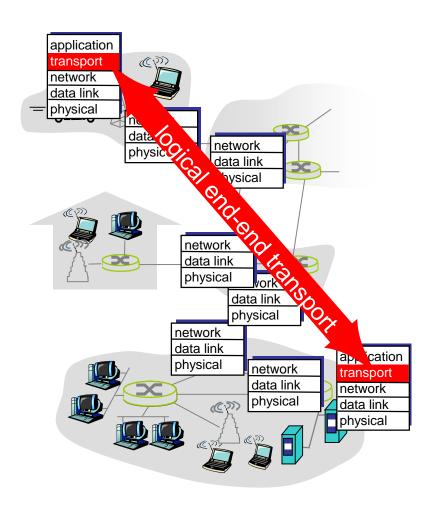
- Communication between processes (e.g., socket = IP address and Port)
- Relies on network layer and serves the application layer
- e.g., TCP and UDP

Network layer

- Logical communication between nodes
- Hides details of the link technology
- e.g., Internet Protocol (IP)

Transport Protocols

- Provide logical communication between application processes running on different hosts
- Run on end hosts
 - Sender: breaks application messages into segments, and passes to network layer
 - Receiver: reassembles segments into messages, passes to application layer
- Multiple transport protocol available to applications
 - Internet: TCP and UDP
 - Other: Datagram Congestion Control Protocol (DCCP), Stream Control Transmission Protocol (SCTP)



Transport vs. network layer

- network layer: logical communication between hosts
- transport layer: logical communication between processes
 - relies on, enhances, network layer services

Household analogy:

- 12 kids sending letters to 12 kids
- app messages = letters in envelopes
- processes = kids
- hosts = houses
- transport protocol = Ann and Bill
- network-layer protocol = postal service

Outline

- Transport-layer services
- Multiplexing and demultiplexing
- Connectionless transport: UDP

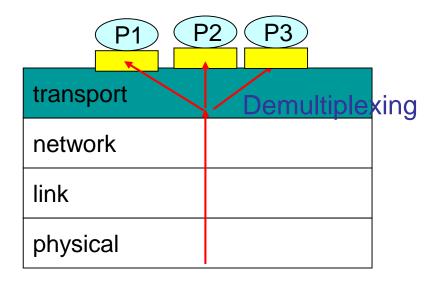
Multiplexing/demultiplexing – Definition

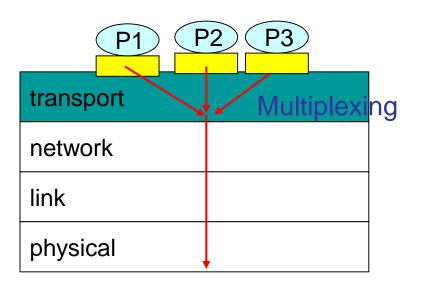
Demultiplexing

 To deliver the data in a transportlayer segment to the correct socket

Multiplexing

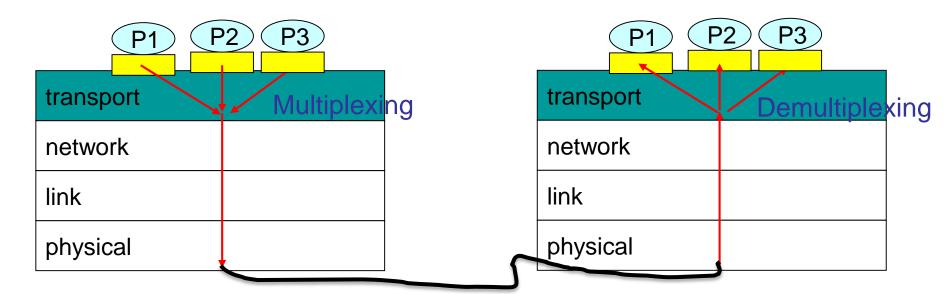
 To create segments and pass them to the network layer.





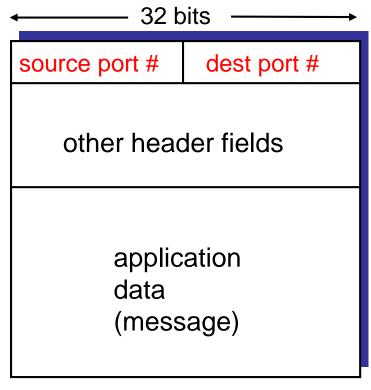
Multiplexing/demultiplexing – Why?

- Why need demultiplexing?
 - Now, the transport layer needs a way to determine which application the packet needs to be delivered. This is the demultiplexing problem.
- Why need multiplexing?
 - To facilitate demultiplexing!



How demultiplexing works

- host receives IP datagrams
 - each datagram has source IP address, destination IP address
 - each datagram carries 1 transport-layer segment
 - each segment has source, destination port number
- host uses IP addresses & port numbers to direct segment to appropriate socket



TCP/UDP segment format

Connectionless demultiplexing

Create sockets with port numbers:

```
DatagramSocket mySocket1 = new
  DatagramSocket(99111);
DatagramSocket mySocket2 = new
  DatagramSocket(99222);
```

UDP socket identified by two-tuple:

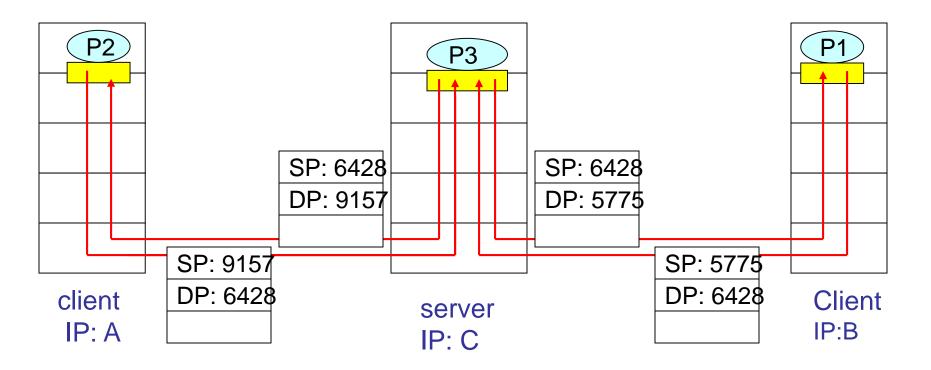
(dest IP address, dest port number)

Why do we need source port #?

- When host receives UDP segment:
 - checks destination port number in segment
 - directs UDP segment to socket with that port number
- IP datagrams with different source IP addresses and/or source port numbers, but with the same destination IP and port number, will be directed to same socket.

Connectionless demux (cont)

DatagramSocket serverSocket = new DatagramSocket(6428);



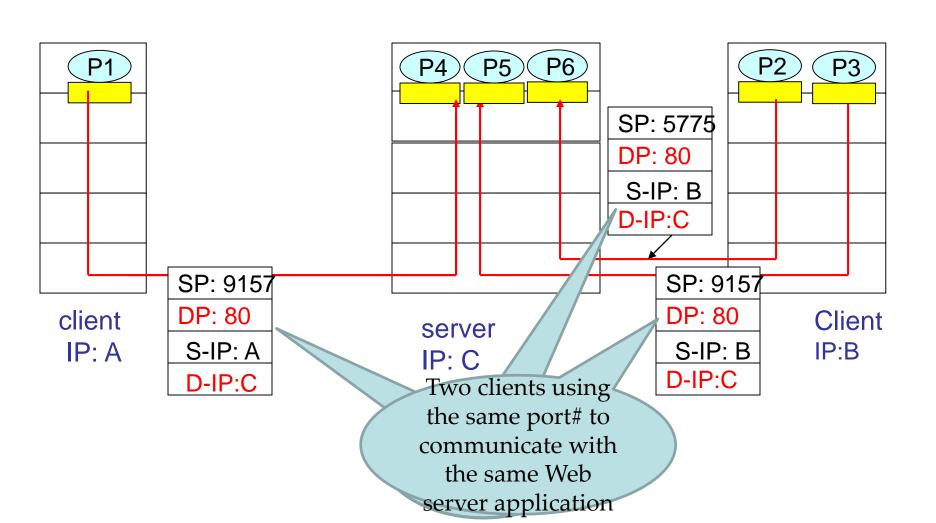
SP provides "return address"

Connection-oriented demux

- TCP socket identified by 4-tuple:
 - source IP address
 - source port number
 - dest IP address
 - dest port number
- recv host uses all four values to direct segment to appropriate socket

- Server host may support many simultaneous TCP sockets:
 - each socket identified by its own 4-tuple
- Web servers have different sockets for each connecting client
 - non-persistent HTTP will have different socket for each request

Connection-oriented demux (cont)



Outline

- Transport-layer services
- Multiplexing and demultiplexing
- Connectionless transport: UDP

UDP: User Datagram Protocol [RFC 768]

- "no frills," "bare bones"Internet transport protocol
- "best effort" service, UDP segments may be:
 - lost
 - delivered out of order to app
- connectionless:
 - no handshaking between UDP sender, receiver
 - each UDP segment handled independently of others

Why is there a UDP?

- no connection establishment (which can add delay)
- simple: no connection state at sender, receiver
- small segment header
- no congestion control: UDP can blast away as fast as desired

UDP: more

 often used for streaming multimedia apps

loss tolerant

rate sensitive

other UDP uses

DNS

SNMP

 reliable transfer over UDP: add reliability at application layer

> application-specific error recovery!

Length, in bytes of UDP segment, including header

→ 32 bits →						
source port #	dest port #					
length	checksum					
Application data (message)						

UDP segment format

Popular Internet Applications and their underlying transport protocols

Application	Application-layer protocol	Underlying transport protocol
Email	SMTP	TCP
Web	HTTP	TCP
Routing protocol	BGP	TCP
Routing protocol	RIP	Typically UDP
Name translation	DNS	Typically UDP
Streaming multimedia	Typically proprietary	Typically UDP
Internet telephony	Typically proprietary	Typically UDP

UDP checksum

Goal: detect "errors" (e.g., flipped bits) in transmitted segment

Sender:

- treat segment contents as sequence of 16-bit integers
- checksum: addition (1's complement sum) of segment contents
- sender puts checksum value into UDP checksum field

Receiver:

- compute checksum of received segment
- check if computed checksum equals checksum field value:
 - NO error detected
 - YES no error detected.
 But maybe errors
 nonetheless? More later

Internet Checksum Example

- Note
 - When adding numbers, a carryout from the most significant bit needs to be added to the result
- Example: add two 16-bit integers

A UDP segment captured by Wireshark

```
12 0.508950
                                  10.34.40.169
                                                                     132.181.2.225
                                                                                                        DNS
      13 0.509041
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                        DNS
      14 0.509462
                                  10.34.40.169
                                                                     132.181.2.225
                                                                                                         DNS
      15 0.510280
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                         DNS
      16 0.510354
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                        DNS
      17 33.503303
                                  10.34.40.169
                                                                     132.181.2.225
                                                                                                        DNS
      18 33.504666
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                         DNS
      19 61.002465
                                  10.34.40.169
                                                                     132.181.2.225
                                                                                                         DNS
      20 61.004045
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                        DNS
      21 61.013117
                                  10.34.40.169
                                                                     132.181.2.225
                                                                                                        DNS
      22 61.014578
                                  132.181.2.225
                                                                     10.34.40.169
                                                                                                        DNS
▶ Frame 12: 88 bytes on wire (704 bits), 88 bytes captured (704 bits) on interface 0
▶ Ethernet II, Src: IntelCor b6:fe:63 (80:19:34:b6:fe:63), Dst: JuniperN ef:61:00 (2c:21:31:ef:61:00)

    Internet Protocol Version 4, Src: 10.34.40.169, Dst: 132.181.2.225

■ User Datagram Protocol, Src Port: 63507, Dst Port: 53
   Source Port: 63507
   Destination Port: 53
   Length: 54
   Checksum: 0xe576 [unverified]
    [Checksum Status: Unverified]
    [Stream index: 6]
Domain Name System (query)
```

Why Would Anyone Use UDP?

- Finer control over what data is sent and when
 - As soon as an application process writes into the socket
 - ... UDP will package the data and send the packet
- No delay for connection establishment
 - UDP just blasts away without any formal preliminaries
 - ... which avoids introducing any unnecessary delays
- No connection state
 - No allocation of buffers, parameters, sequence #s, etc.
 - ... making it easier to handle many active clients at once
- Small packet header overhead
 - UDP header is only eight-bytes long

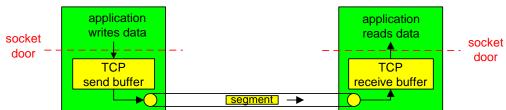
Outline

- Connection-oriented transport: TCP
 - Overview and segment structure
 - Connection management
 - Reliable data transfer
 - Flow control

TCP: Overview

RFCs: 793, 1122, 1323, 2018, 2581

- point-to-point:
 - one sender, one receiver
- reliable, in-order byte steam:
 - no "message boundaries"
- pipelined:
 - TCP congestion and flow control set window size
- send & receive buffers



full duplex data:

- bi-directional data flow in same connection
- MSS: maximum segment size

connection-oriented:

 handshaking (exchange of control msgs) init's sender, receiver state before data exchange

flow controlled:

 sender will not overwhelm receiver

TCP segment structure

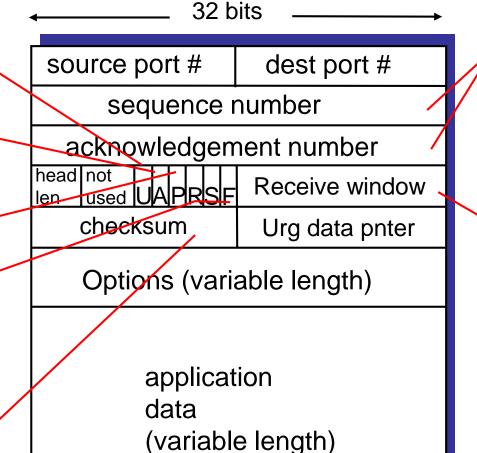
URG: urgent data (generally not used)

ACK: ACK # valid

PSH: push data now (generally not used)

RST, SYN, FIN: connection estab (setup, teardown commands)

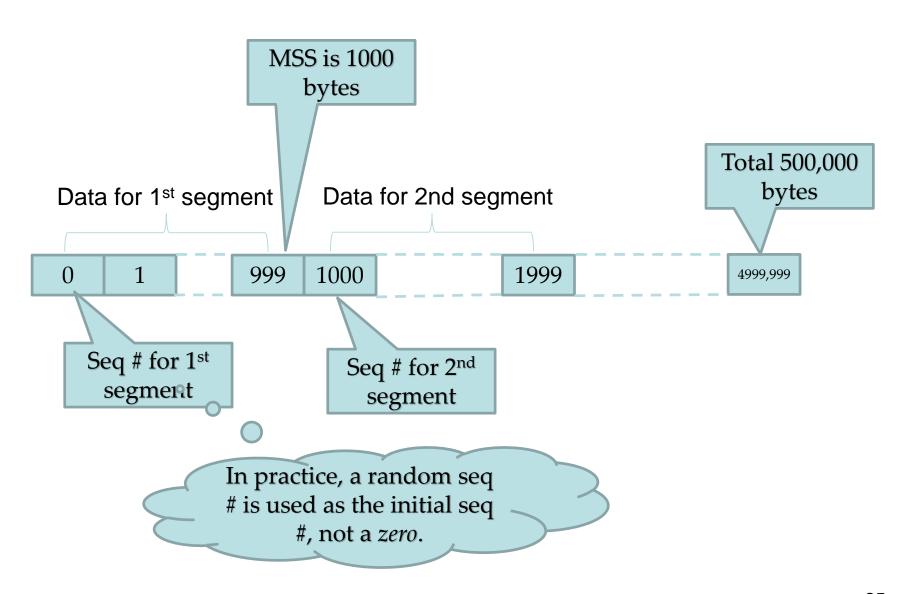
> Internet checksum' (as in UDP)



counting by bytes of data (not segments!)

bytes
rcvr willing
to accept

TCP segments



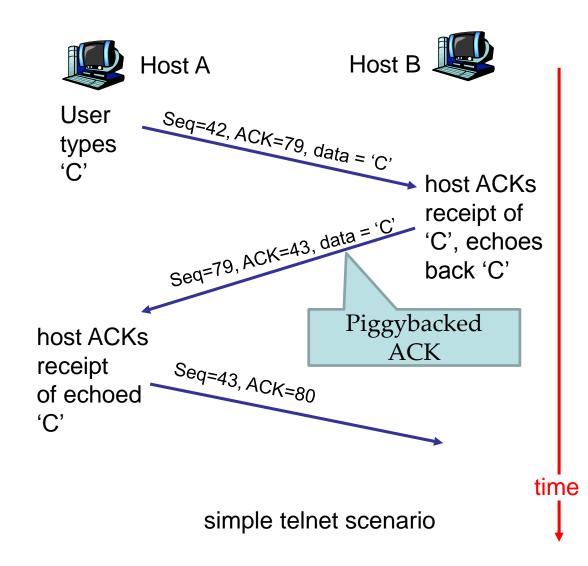
TCP seq. #'s and ACKs

Seq. #'s:

byte stream
 "number" of first
 byte in segment's
 data

ACKs:

- seq # of next byte expected from other side
- cumulative ACK
- Q: how receiver handles out-of-order segments
 - A: TCP spec doesn't say, - up to implementor



TCP Connection Management

Three way handshake:

Step 1: client host sends TCP SYN segment to server

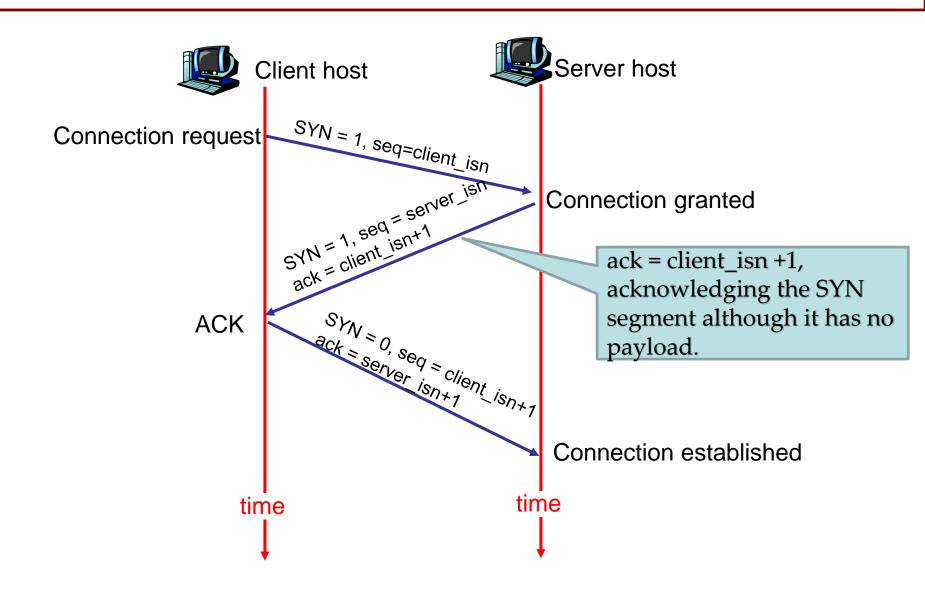
- specifies initial seq #
- no data

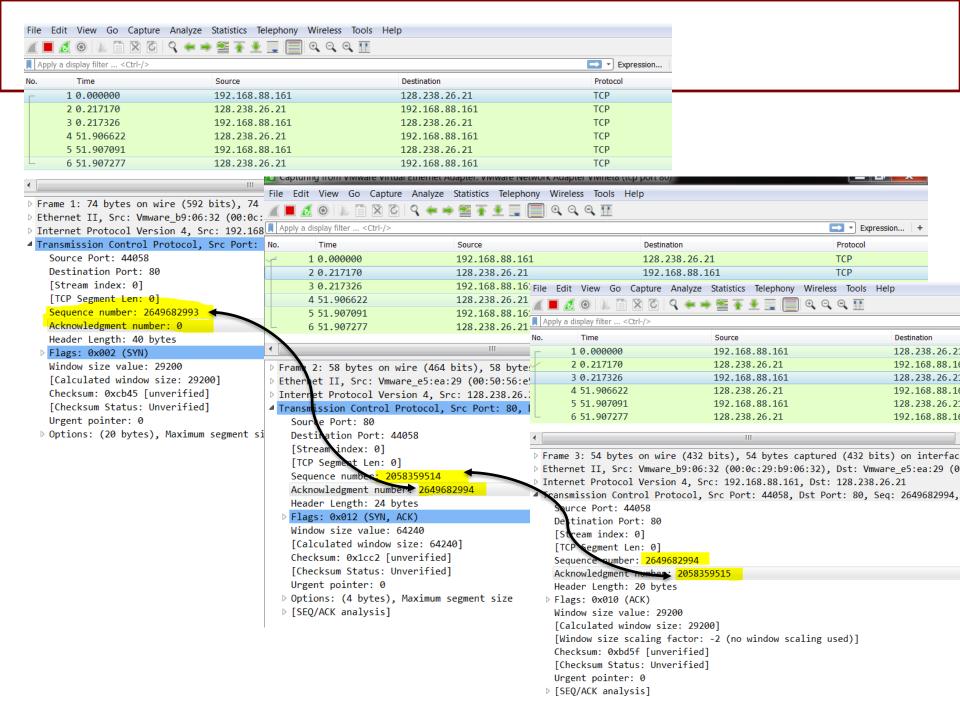
Step 2: server host receives SYN, replies with SYNACK segment

- server allocates buffers
- specifies server initial seq. #

Step 3: client receives SYNACK, replies with ACK segment, which may contain data

TCP 3-way handshaking





TCP Connection Management (closing)

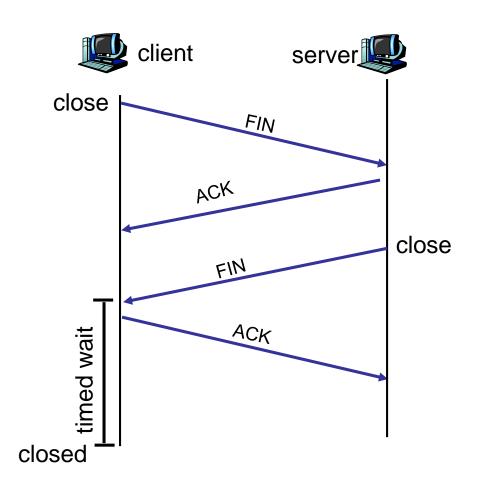
All good things must come to an end:

Step 1: client end system sends TCP FIN control segment to server

Step 2: server receives FIN, replies with ACK. Closes connection, sends FIN.

Step 3: client receives FIN, replies with ACK.

Step 4: server, receives ACK. Connection closed.



			30 100.493931	132.101.107.23	10.34.40.109		
			39 160.496329	132.181.107.25	10.34.40.169		
		_	40 160.496413	10.34.40.169	132.181.107.25		
37 160.494102	10.34.40.169	132.181.107.25	Frame 40: 54 bytes on wi	re (432 hits) 54 hytes cantu	red (432 hits) on interface 0		
38 160.495931	132.181.107.25	10.34.40.169	→ Frame 40: 54 bytes on wire (432 bits), 54 bytes captured (432 bits) on interface 0 ▶ Ethernet II, Src: IntelCor b6:fe:63 (80:19:34:b6:fe:63), Dst: JuniperN ef:61:00 (2c:21:31:				
39 160.496329	132.181.107.25	10.54.40.105	> Internet Protocol Version 4, Src: 10.34.40.169, Dst: 132.181.107.25				
40 160.496413	10.34.40.169						
Tatawat Dartson Ware	- 4 C 10 24 40 100			tocol, Src Port: 5/483, Dst P	ort: 25, Seq: 3400705034, Ack: 154504		
Internet Protocol Version		DST: 132.181.107.25 DST PORT: 25, Seq: 3400705033, Ack: 1545042735,	Source Port: 57483				
Source Port: 57483	toco1, 3rc Port. 37483,	DSC POPC. 25, 3eq. 5400705055, ACR. 1545042755,	Destination Port: 25				
Destination Port: 25		37 160.494102 10.34	[Stream index: 1]	10			
[Stream index: 1]			[TCP Segment Len: 0]	ACI	(
[TCP Segment Len: 0]			Sequence number: 34007	705034			
Sequence number: 340070	05033	39 160.496329 132.1	Acknowledgment number:				
Acknowledgment number:	1545042735	40 160.496413 10.34	Header Length: 20 byte				
Header Length: 20 byte:	5	•		-5			
◆ Flags: 0x011 (FIN, ACK))	Pame 38: 54 bytes on wire (432 bits)	▲ Flags: 0x010 (ACK)				
000 = Res		Ethernet II, Src: JuniperN_ef:61:00 (000 = Res				
0 = Non			0 = Nor	nce: Not set			
		Internet Protocol Version 4, Src: 132	0 = Cor	ngestion Window Reduced (CWR):	Not set		
0 = ECN		▲ Transmission Control Protocol, Src Po	0 = ECN	N-Echo: Not set			
0 = Urg 1 = Ack		Source Port: 25	0 = Urg	gent: Not set			
0 = Pus	-	Destination Port: 57483	= Ack	knowledgment: Set			
0 = Res		[Stream index: 1]	0 = Pus				
0. = Syn		[TCP Segment Len: 0]	0 = Res				
▷ 1 = Fin		Sequence number: 1545042735					
[TCP Flags: ·····A			0. = Syr				
Window size value: 256		Acknowledgment number 3400705034	0 = Fir				
[Calculated window size	e: 65536]	Header Length: 20 bytes	[TCP Flags: ·····/	•			
[Window size scaling fa	actor: 256]	◆ Flags: 0x010 (ACK)	Window size value: 256	5			
Checksum: 0x6592 [unver		000 = Reserved: Not sε	[Calculated window siz	ze: 65536]			
[Checksum Status: Unver	rified]	0 = Nonce: Not set	[Window size scaling f	factor: 256]			
Urgent pointer: 0		0 = Congestion Windo	Checksum: 0x6591 [unve				
EIA.		0 = ECN-Echo: Not set	[ICP Segment Ler				
FIN			Sequence number:				
		0 = Urgent: Not set		number: 3400705034			
		1 = Acknowledgment: Se					
		0 = Push: Not set	▲ Flags: 0x011 (F)				
		0 = Reset: Not set		. = Reserved: Not set			
		0. = Syn: Not set		. = Nonce: Not set			
		0 = Fin: Not set		. = Congestion Window Reduced (Cl	√R): Not set		
		[TCP Flags: ······A····]		. = ECN-Echo: Not set			
		Window size value: 513		. = Urgent: Not set			
				. = Acknowledgment: Set			
		[Calculated window size: 131328]		. = Push: Not set			
		[Window size scaling factor: 256]		. = Reset: Not set			
		Checksum: 0x6491 [unverified]	and the second s	. = Syn: Not set			
			Þ				
		.,	[TCP Flags: ·	-			
		ACK	Window size valu				
		7.0	[Calculated wind	dow size: [131328]			
				aling factor 250]			
			Chackeum: 0v6/00	A mylenitied I			

Checksum: 0x6490 [unverified]

37 160.494102

38 160.495931

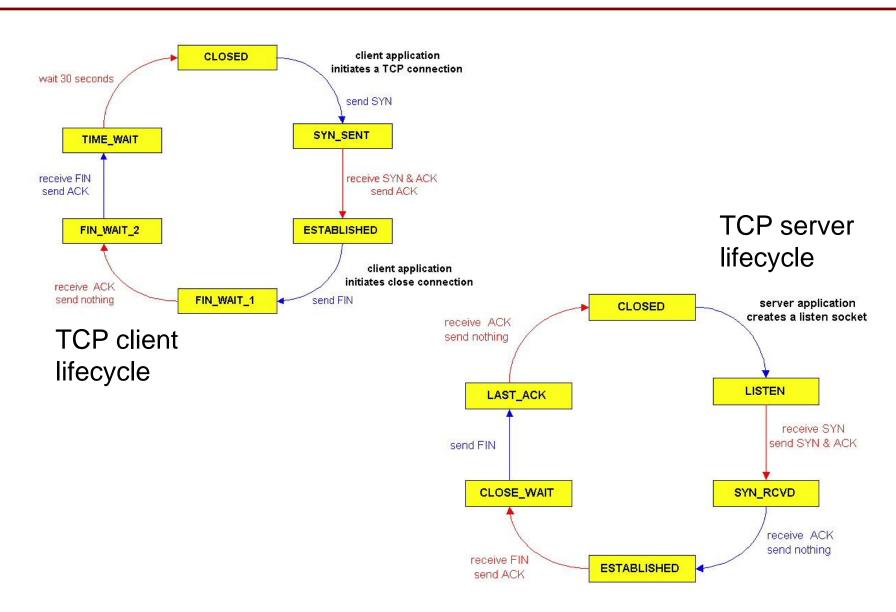
10.34.40.169

132.181.107.25

132.181.107.25

10.34.40.169

TCP Connection Management (cont)



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 - Reliable data transfer
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References

- [KR3] James F. Kurose, Keith W. Ross, Computer networking: a top-down approach featuring the Internet, 3rd edition.
- [PD5] Larry L. Peterson, Bruce S. Davie, Computer networks: a systems approach, 5th edition
- [TW5] Andrew S. Tanenbaum, David J. Wetherall, Computer network, 5th edition
- [LHBi]Y-D. Lin, R-H. Hwang, F. Baker, Computer network: an open source approach, International edition

Acknowledgements

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 - Dr DongSeong Kim's slides for COSC264, University of Canterbury;
 - Prof Aleksandar Kuzmanovic's lecture notes for CS340, Northwestern University, https://users.cs.northwestern.edu/~akuzma/class

es/CS340-w05/lecture_notes.htm