

Introduction to Transport Layer

Lecture 6 | Part 1 | CSE421 - Computer Networks

Department of Computer Science and Engineering School of Data & Science



Objectives

our goals:

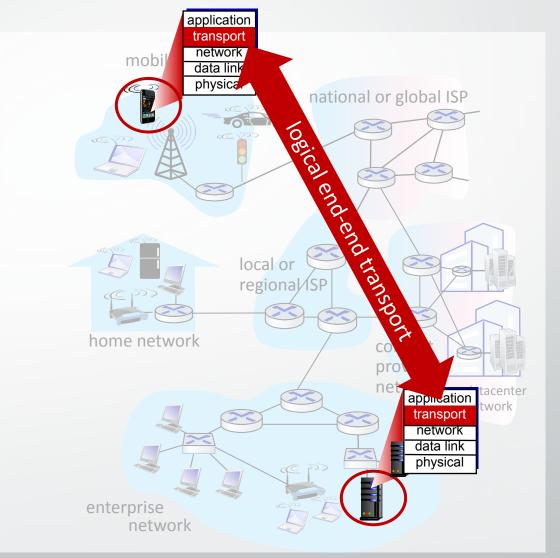
- understand principles behind transport layer services:
 - reliable data transfer, segmentation, flow control etc..

- learn about Internet transport layer protocols:
 - UDP: connectionless transport
 - TCP: connection-oriented reliable transport

Transport services and protocols

- provide *logical communication* between application processes running on different hosts
- transport protocols actions in end systems:
 - sender: breaks application messages into segments, passes to network layer
 - receiver: reassembles segments into messages, passes to application layer
- two transport protocols available to Internet applications

TCP, UDP



Transport vs. Network layer

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

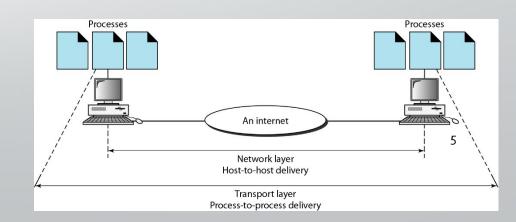
household analogy:

- 12 kids in Ann's house sending letters to 12 kids in Bill's house:
- hosts = houses
- processes = kids
- app messages = letters in envelopes
- transport protocol = Ann and Bill who demux to in-house siblings
- network-layer protocol = postal service





- The transport layer is responsible for the delivery of a message from one process (sender) to another (receiver).
- Transport Layer PDU is called Segments
- Functions:
 - Segmentation and Reassembly
 - Adds Port Address to identify the application
 - Multiplexing
 - Connection establishment and termination
 - Flow and Error Control



Purpose of the Transport Layer Primary responsibilities:

- Segmenting the data and managing each piece.
- Reassembling the segments into streams of application data.
- Identifying the different applications.
- Multiplexing
- Initiating Reliability on.
- Performing flow control between end users.
- Enabling error recovery.

Reliability

- •Three basic operations of reliability are:
 - Initiating sessions and tracking transmitted data
 - Acknowledging received data
 - Retransmitting any unacknowledged data
- •To support these reliability operations, more control data is added in the Layer 4 header.

Different Applications

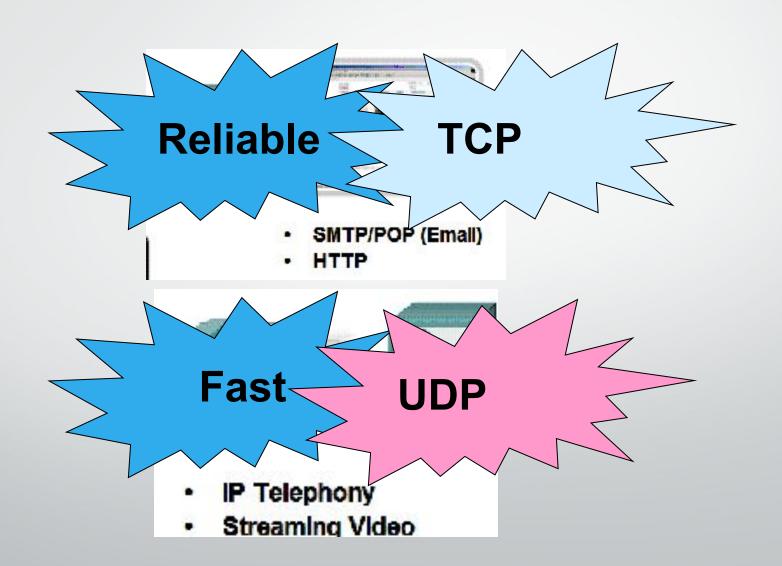
Different Requirements



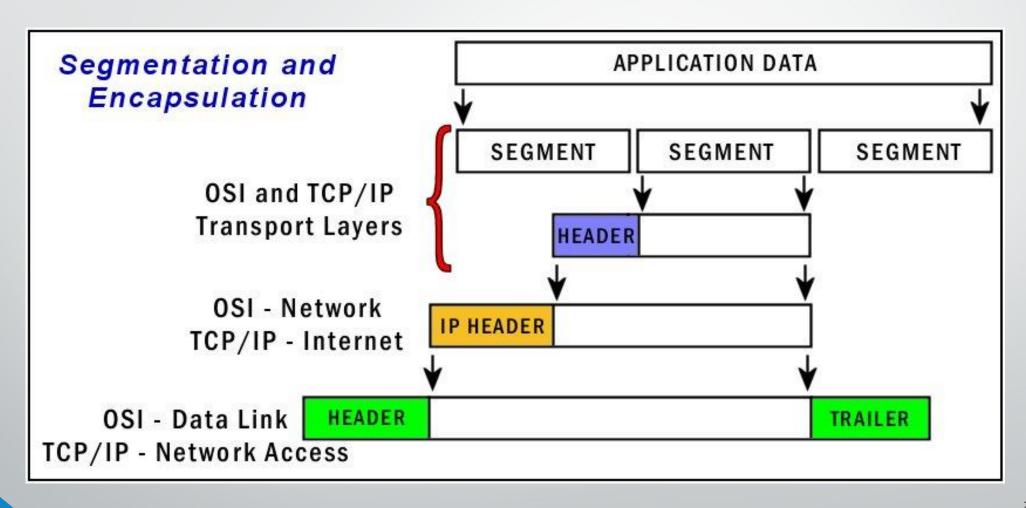


- Some applications need their data to be complete with no errors or gaps and they can accept a slight delay to ensure this.
- Some applications can accept occasional errors or gaps in the data but they cannot accept any delay.

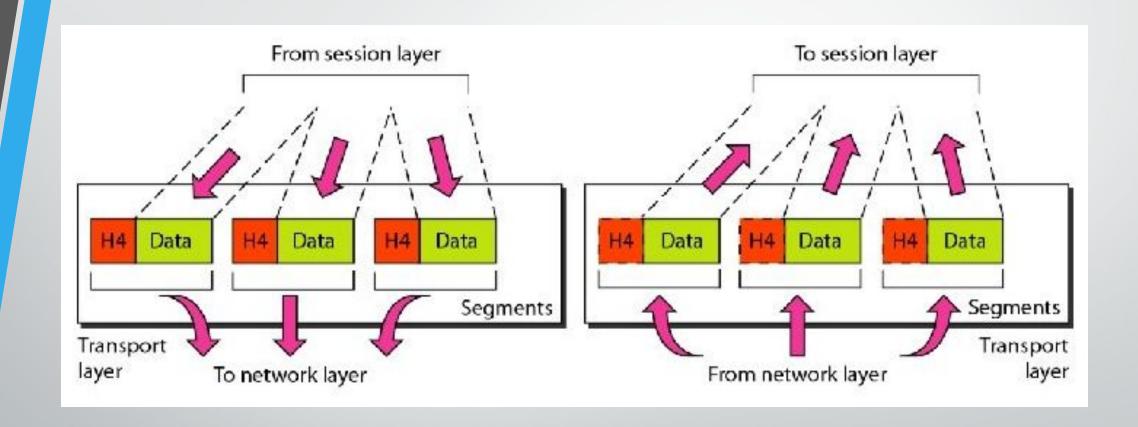
Solution: Two transport protocols?



Function 1 – Segmentation and Reassembly

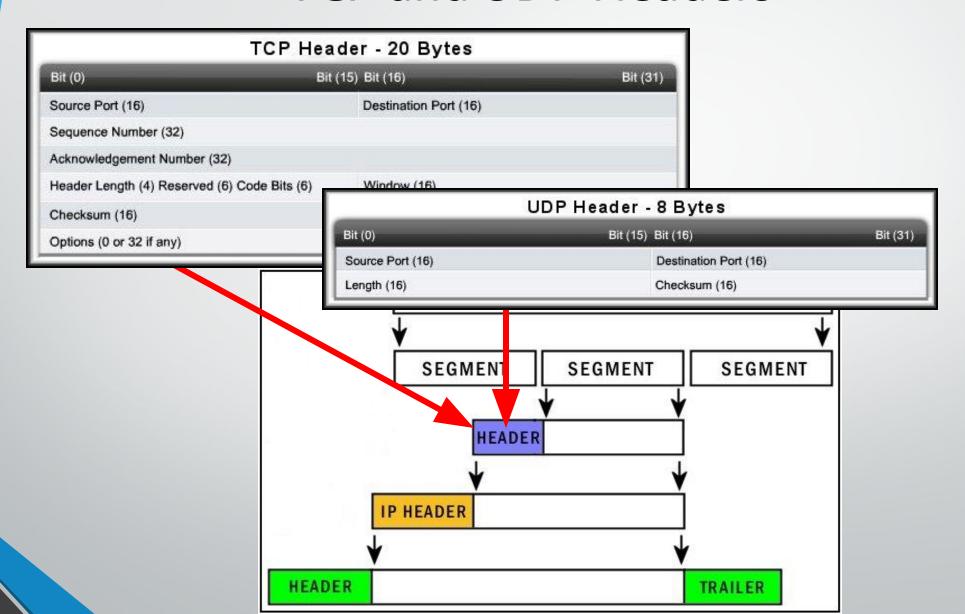


Function 1 – Segmentation and Reassembly

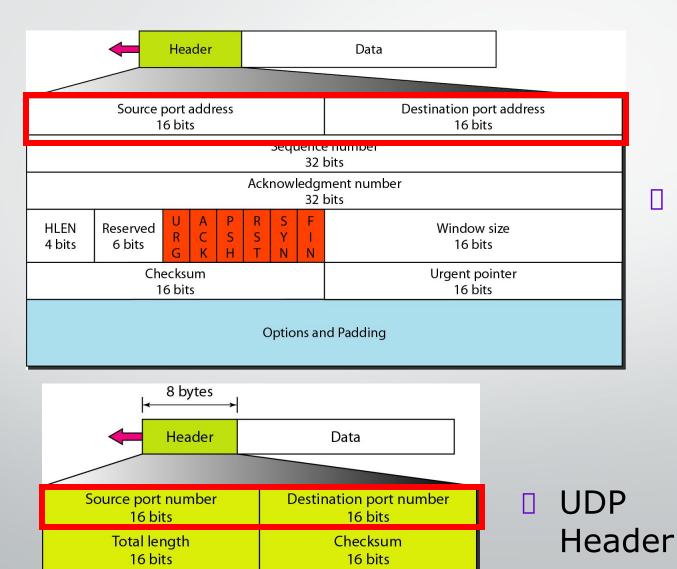


Also known as encapsulation and de-capsulation.

TCP and UDP Headers



TCP and UDP Headers

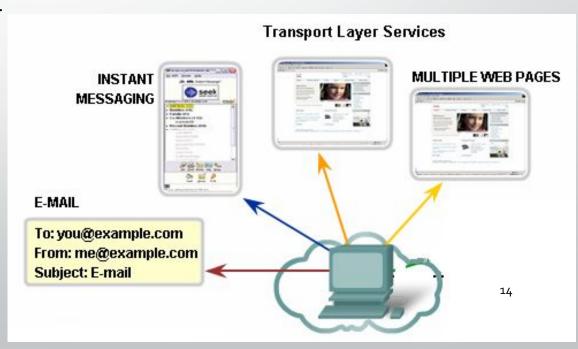


TCPHeader

Function 2 – Identification Applications Using Port Address

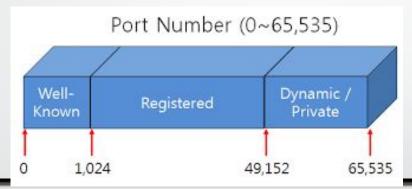


- Port Numbers/Addresses are used to identify different applications/processes running in a computer
- 16-bit in length
 - Represented as one single decimal number
 - Range **o 65535**
 - e.g. 80 Web; 25 SMTP
 - 110 POP3, 531 Instant Messaging



Port Numbers

- Internet Corporation for Assigned Names and Numbers (ICANN) assigns port numbers.
- Three categories:



Port Number Range	Port Group
0 to 1023	Well Known (Contact) Ports
1024 to 49151	Registered Ports
49152 to 65535	Private and/or Dynamic Ports

Port Addressing Types

- Well-Known Ports:
 - Reserved for common services and applications.

Pre-assigned by ICANN

Port Number Range	Port Group
0 to 1023	Well Known (Contact) Ports
1024 10 49 15 1	Registered Ports
49152 to 65535	Private and/or Dynamic Ports

20 – FTP Data

25 – SMTP

443 – HTTPS

21 – FTP Control

110 – POP3

69 – TFTP

23 – Telnet

520 - RIP

Port Addressing Types

• Registered Ports:

 Port numbers that companies and other users register with ICANN for use by the applications that communicate using any one of the transport layer protocols.

ort Number Range	Port Group
0 to 1023	Well Known (Contact) Ports
1024 to 49151	Registered Ports
49152 to 65535	Private and/or Dynamic Ports

8008 – Alternate HTTP

1863 – MSN Messenger

8080 – Alternate HTTP

5004 - RTP

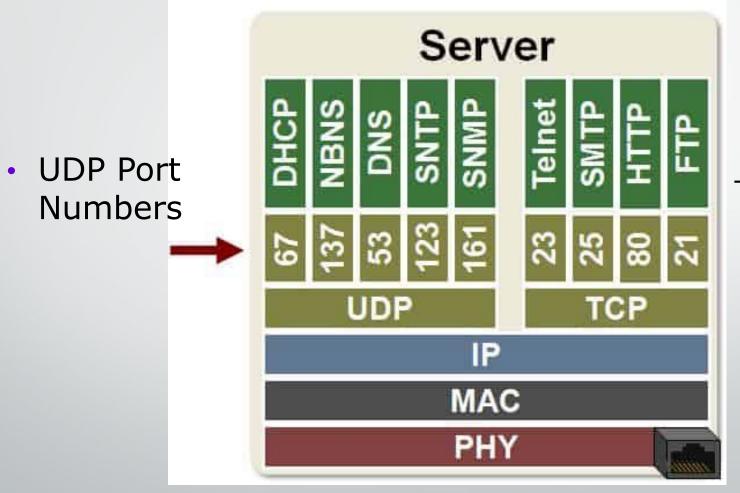
5060 - SIP (VoIP)

Port Addressing Types

- Dynamic Ports:
 - Assigned to a user application at connect time.

Port Number Range	Port Group
0 to 1023	Well Known (Contact) Ports
1024 to 49151	Registered Ports
49152 to 65535	Private and/or Dynamic Ports

Port Numbers

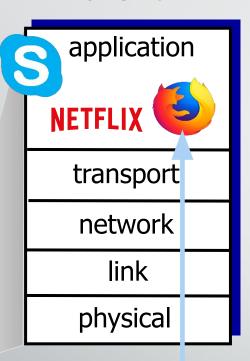


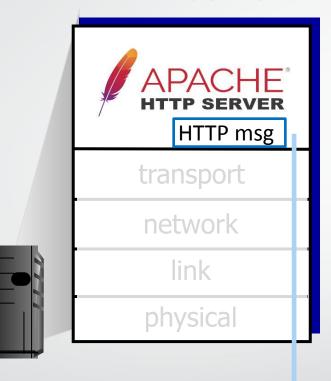
TCP Port Numbers

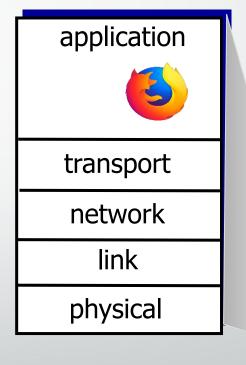
Function 3 - Multiplexing

HTTP server

client



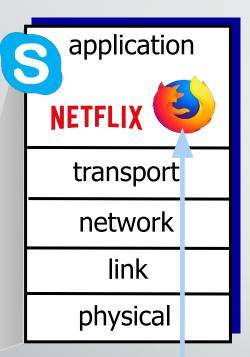


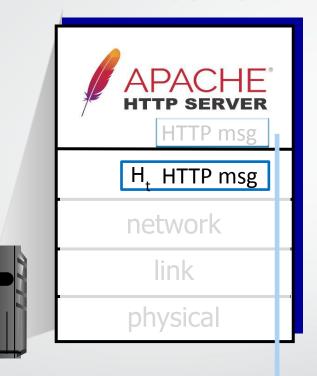


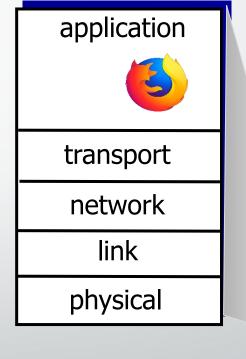


HTTP server





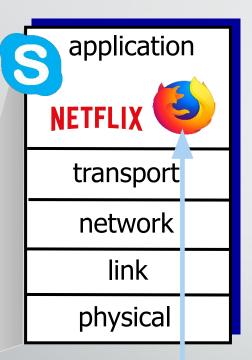


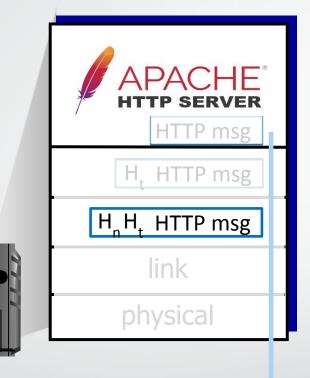


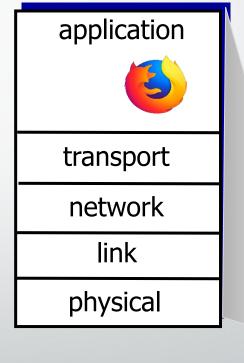


HTTP server



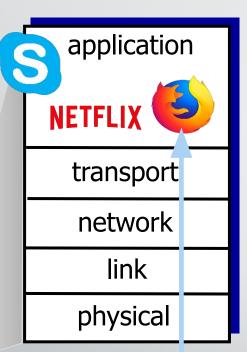






HTTP server





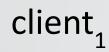


application
transport
network
link
physical



H_nH_t HTTP msg

HTT





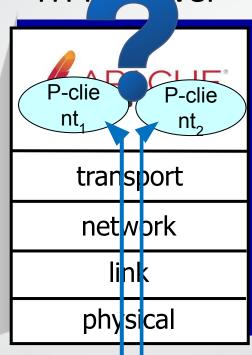
transport

network

link

physical





client₂

application



transport

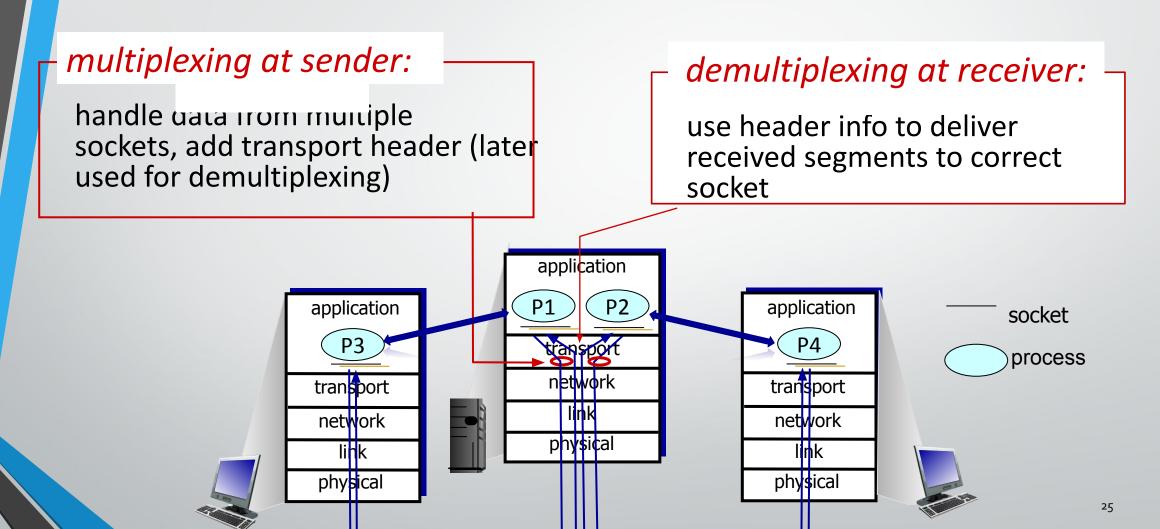
network

link

physical

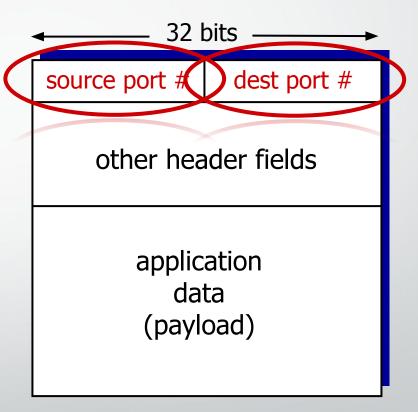


Multiplexing/demultiplexing



How demultiplexing works

- host receives IP datagrams
 - each datagram has source IP address, destination IP address
 - each datagram carries one 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

- when creating datagram to send into UDP socket, must specify
 - destination IP address
 - destination port #

when receiving host receives *UDP* segment:

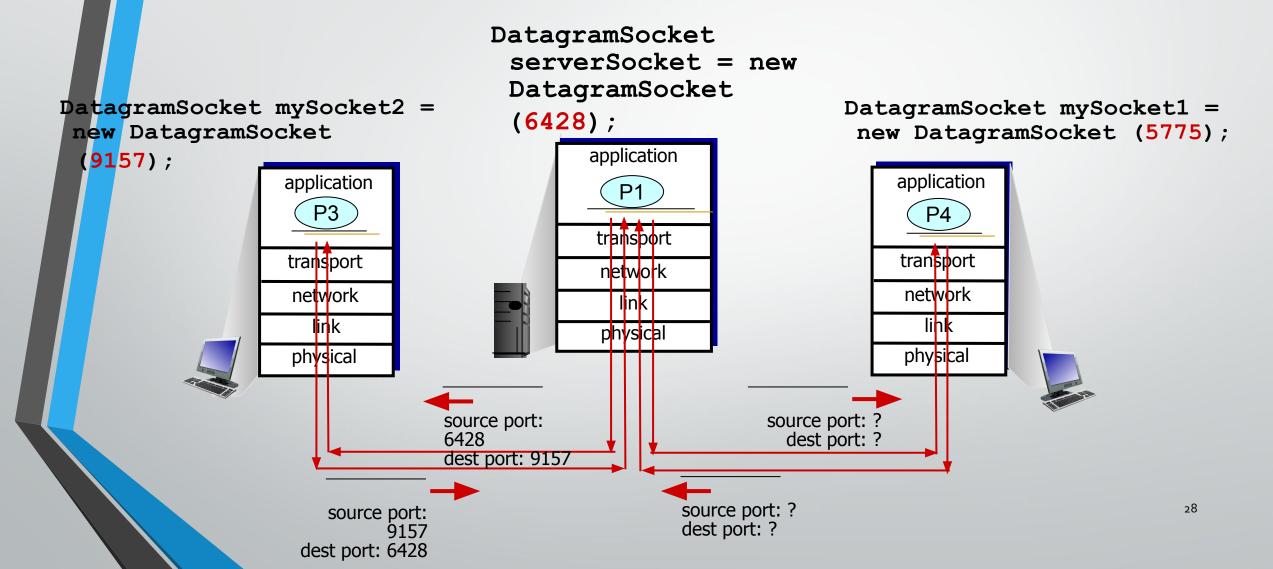
- checks destination port # in segment
- directs UDP segment to socket with that port #



IP/UDP datagrams with same dest.

port #, but different source IP
addresses and/or source port
numbers will be directed to same
socket at receiving host

Connectionless demultiplexing: an example

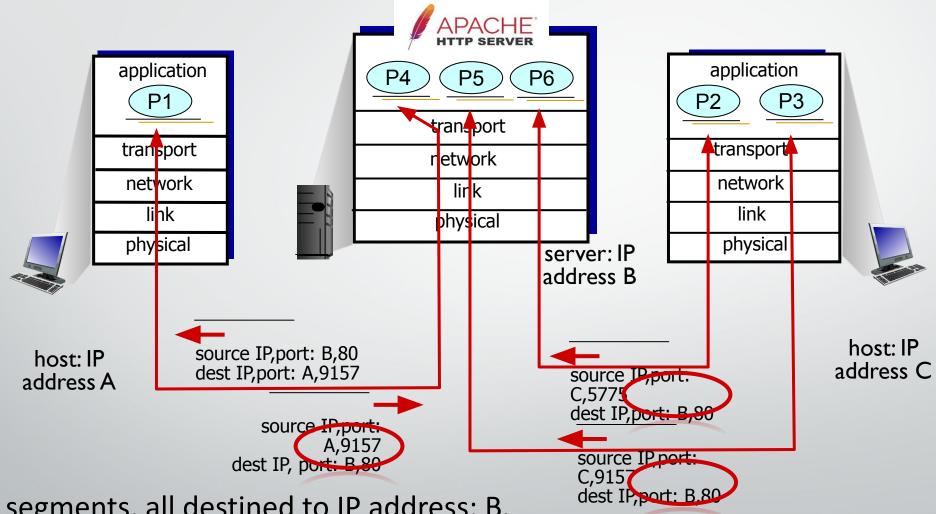


Connection-oriented demultiplexing

- TCP socket identified by 4-tuple:
 - source IP address
 - source port number
 - dest IP address
 - dest port number
- demux: receiver uses all four values (4-tuple) to direct segment to
 appropriate socket

- server may support many simultaneous TCP sockets:
 - each socket identified by its own 4-tuple
 - each socket associated with a different connecting client

Connection-oriented demultiplexing: example



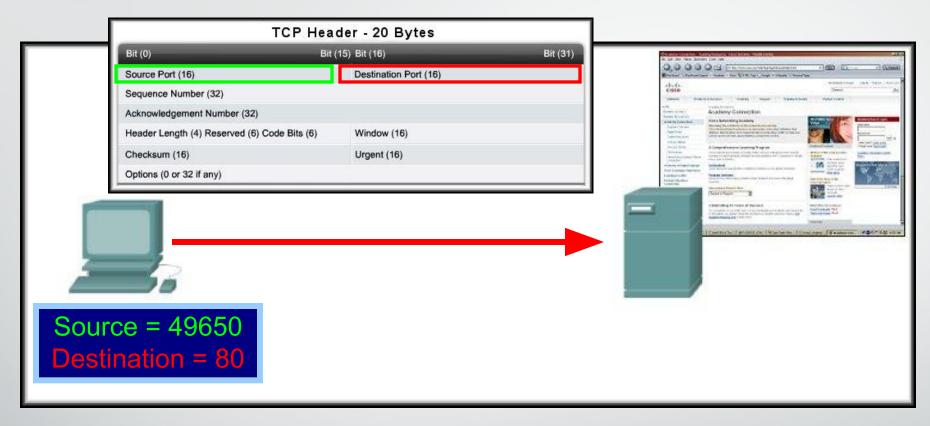
Three segments, all destined to IP address: B,

dest port: 80 are demultiplexed to different sockets

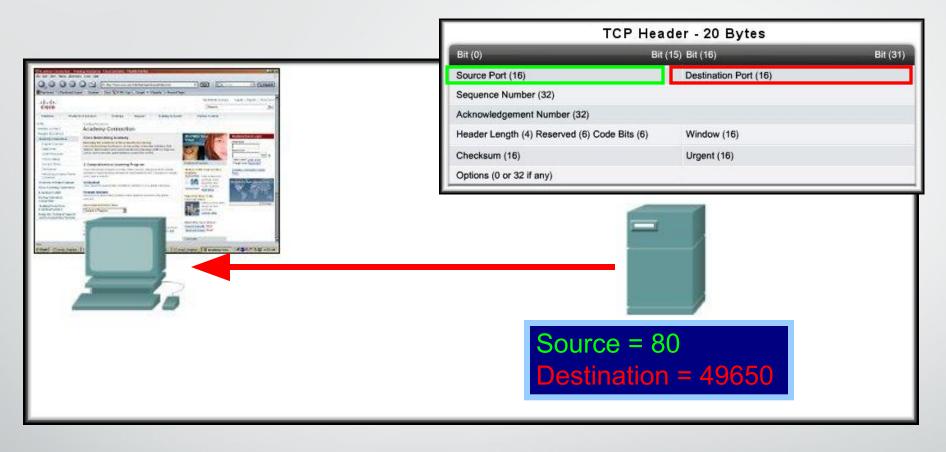
Summary

- Multiplexing, demultiplexing: based on segment, datagram header field values
- UDP: demultiplexing using destination port number (only)
- TCP: demultiplexing using 4-tuple: source and destination IP addresses, and port numbers
- Multiplexing/demultiplexing happen at all layers

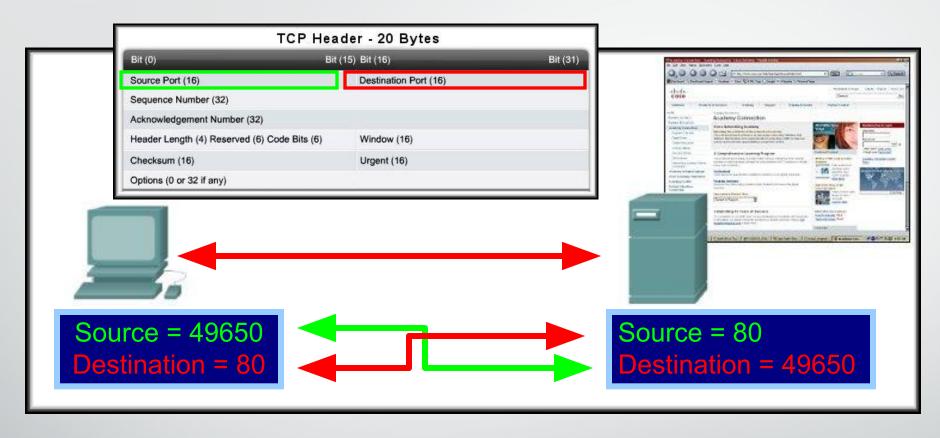
More on Port Numbers



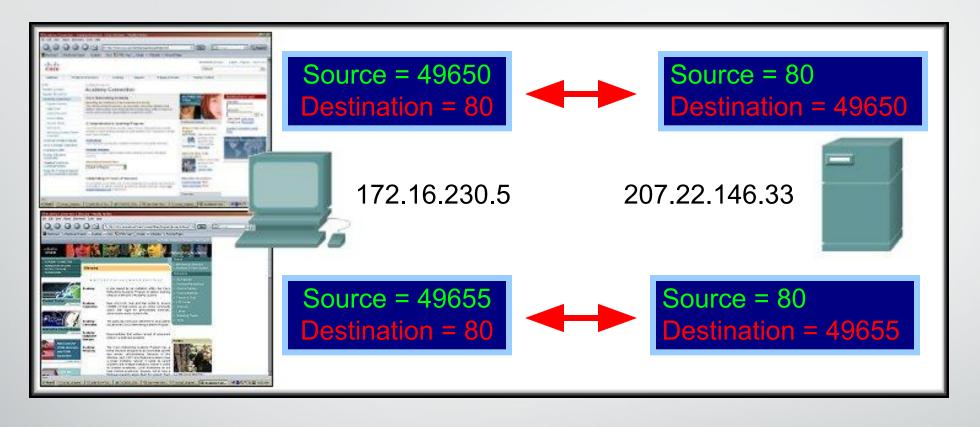
- Server is listening on Port 8o for HTTP connections.
- The client sets the destination port to 80 and uses a dynamic port as its source.



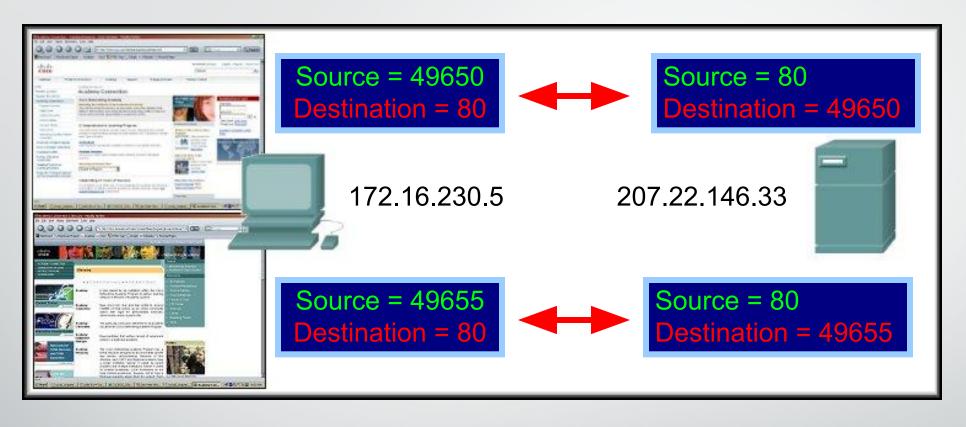
- Server replies with the web page.
 - Sets the source port to 80 and uses the client's source port as the destination.



- Notice how the source and destination ports are used.
- Clients can use any random port number, servers can't.
 - Because clients won't be able to identify server process otherwise
- Servers, however, cannot use any random port number
 - Use of well-known port numbers!

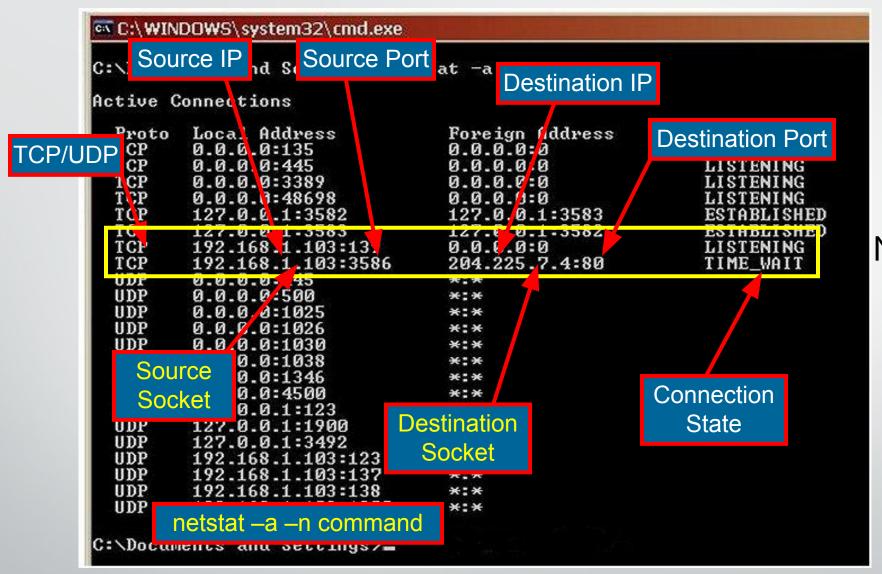


- What if there are two sessions to the same server?
 - The client uses another dynamic port as its source and the destination is still port 80.
 - Different source ports keep the sessions unique on the server.



- How does the Transport Layer keep them separate?
 - The socket (IP Address:Port)

172.16.230.5:49650 172.16.230.5:49655 207.22.146.33:49655



Netstat -Network Utility Tool

 Actually, when you open up a single web page, there are usually several TCP sessions created, not just one.

UDP: User Datagram Protocol [RFC 768]



UDP is a <u>connectionless</u>, <u>unreliable</u> protocol that has <u>no</u> flow and error control. It uses port numbers to track and multiplex data received from the application layer.

why is there a UDP?

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

UDP

- UDP use:
 - streaming multimedia apps (loss tolerant, rate sensitive)
 - DNS
 - SNMP
 - HTTP/3
- reliable transfer over UDP:
 - add reliability at application layer
 - application-specific error recovery!



User Datagram Protocol (UDP)

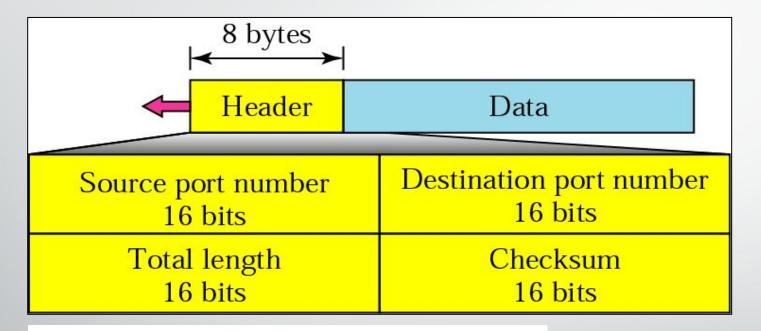


Figure 14: User Datagram Header format

- Connectionless
- No reassembly to order.
- No Error checking
- No Flow control

UDP checksum

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

	1 st number	2 nd number	sum	
Transmitted:	5	6	11	
Received:	4	6	11	
	receiver-	computed 🗲		
		checksum	checksum (as received)	

UDP checksum

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

sender:

- treat contents of UDP segment (including UDP header fields and IP addresses) as sequence of 16-bit integers
- checksum: addition (one's complement sum) of segment content
- checksum value put intoUDP checksum field

receiver:

- compute checksum of received segment
- check if computed checksum equals checksum field value:
 - Not equal error detected
 - Equal no error detected. But maybe errors nonetheless? More later

Internet checksum: an example

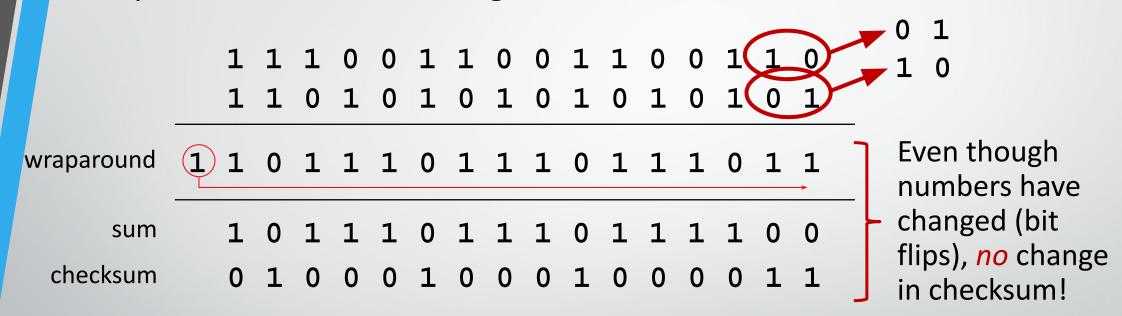
example: add two 16-bit integers

		1	1	1	0	0	1	1	0	0	1	1	0	0	1	1	0
		1	1	0	1	0	1	0	1	0	1	0	1	0	1	0	1
wraparound	1	1	0	1	1	1	0	1	1	1	0	1	1	1	0	1	1
sum		1	0	1	1	1	0	1	1	1	0	1	1	1	1	0	0
checksum		0	1	0	0	0	1	0	0	0	1	0	0	0	0	1	1

Note: when adding numbers, a carryout from the most significant bit needs to be added to the result

Internet checksum: weak protection!

example: add two 16-bit integers



THE END