

# Introduction to Transport Layer

Lecture 6 | Part 1 | CSE421 – Computer Networks

Department of Computer Science and Engineering School of Data & Science



### Objectives

# our goals:

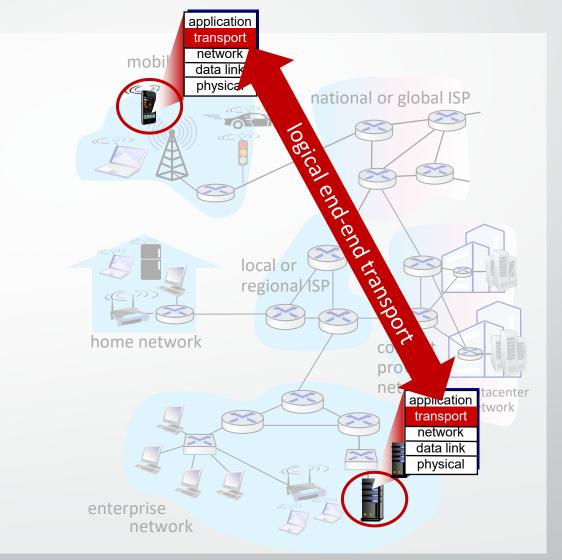
- understand principles behind transport layer services:
  - reliable data transfer, segmentation, flow control etc..

- 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

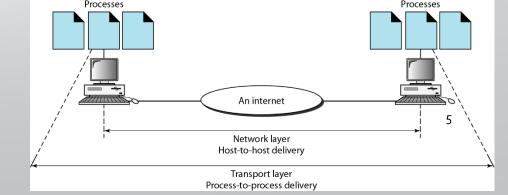
# household analogy:

- 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

### Transport Layer



- 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



\*PDU – Protocol Data Unit

## 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
  - Initiatind on.
  - Reliability Perform n 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.

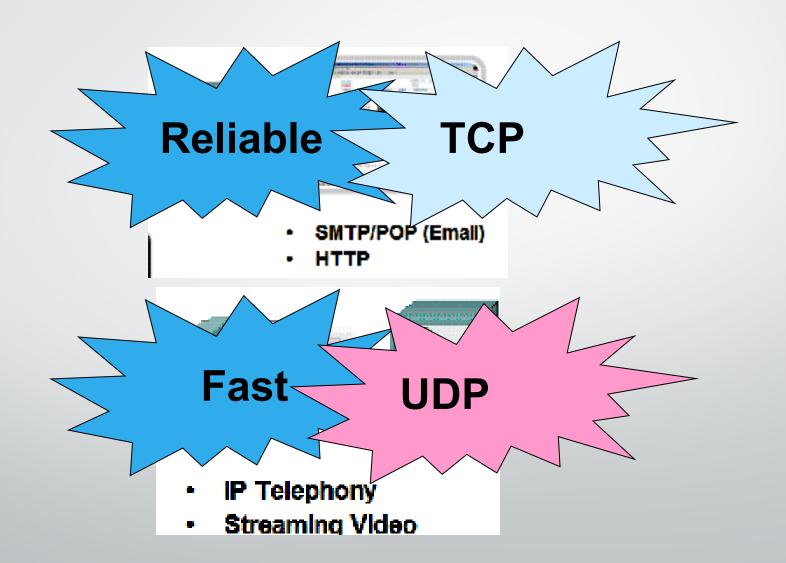
# Different Applications Different Requirements



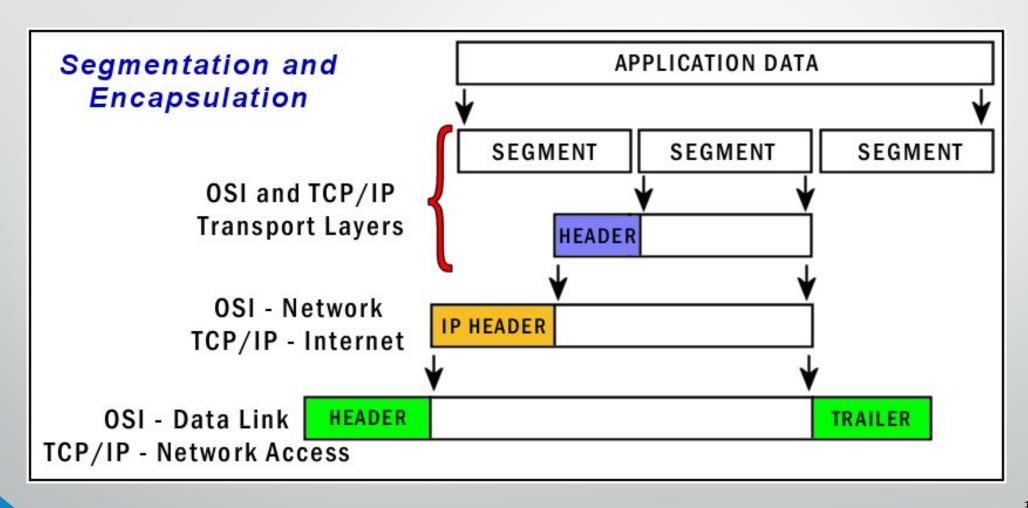


- Some applications need their data to be complete with no errors or gaps and they can accept a slight delay to ensure this.
- Some applications can accept occasional errors or gaps in the data but they cannot accept any delay.

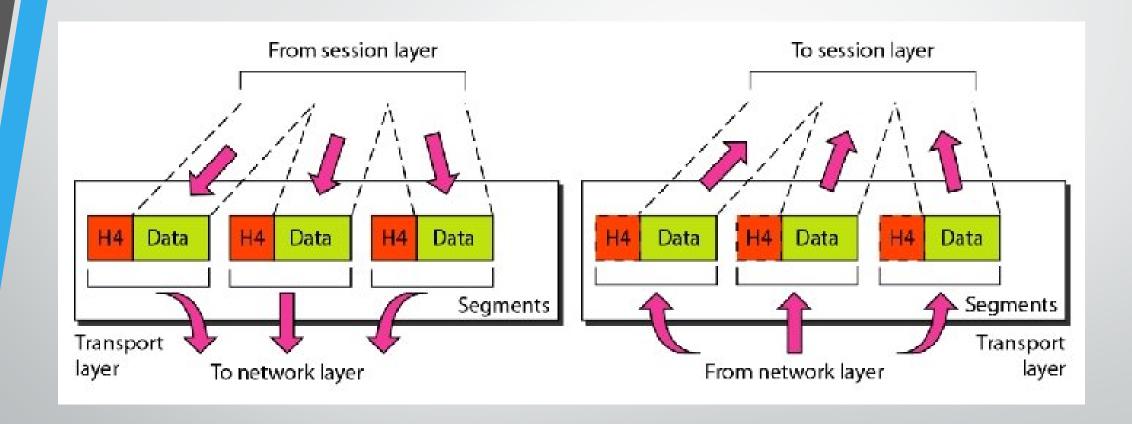
# Solution: Two transport protocols?



# Function 1 – Segmentation and Reassembly

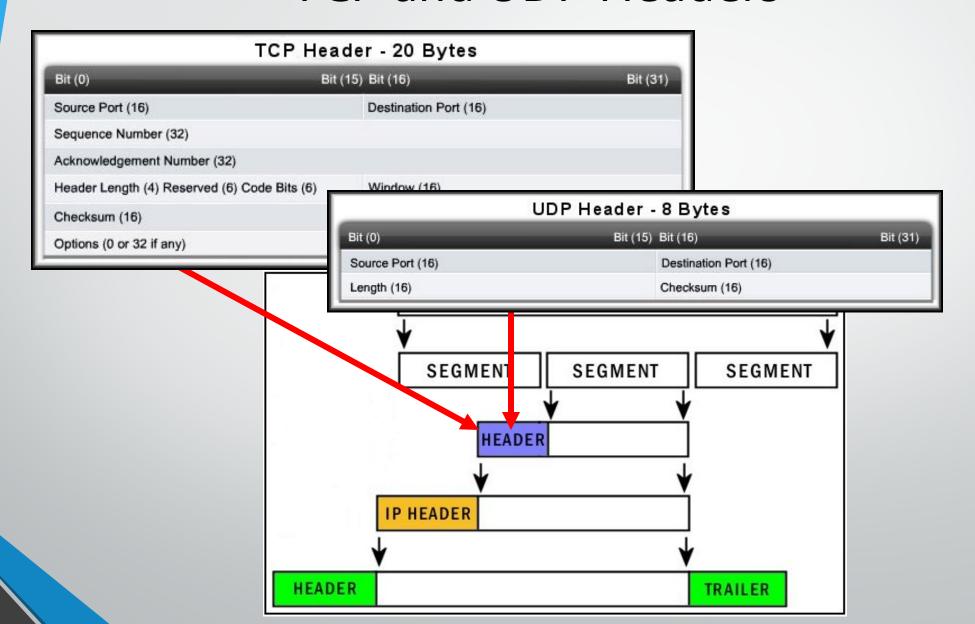


## Function 1 – Segmentation and Reassembly

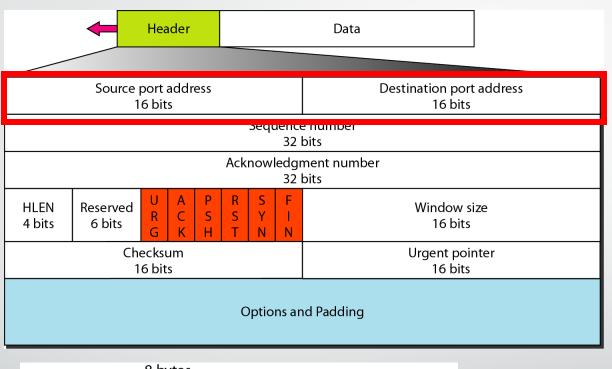


Also known as encapsulation and de-capsulation.

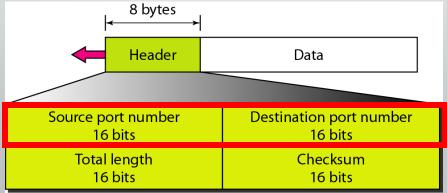
### TCP and UDP Headers



### TCP and UDP Headers



TCP
Header

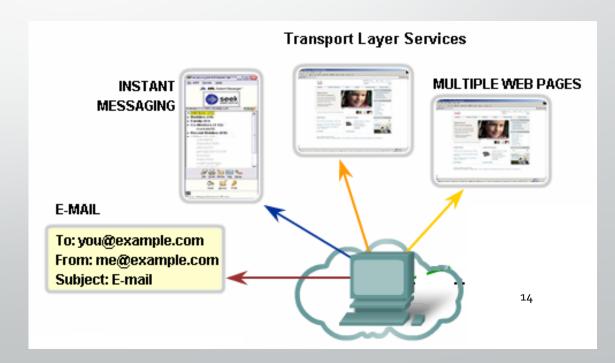


UDP
Header

# Function2 – Identification Applications Using Port Address

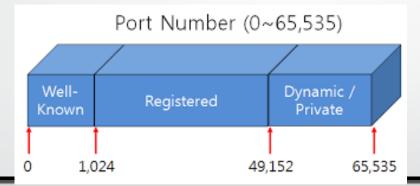


- Port Numbers/Addresses are used to identify different applications/processes running in a computer
- 16-bit in length
  - Represented as one single decimal number
  - Range **o 65535**
  - e.g. 80 Web; 25 SMTP
  - 110 POP3, 531 Instant Messaging



### **Port Numbers**

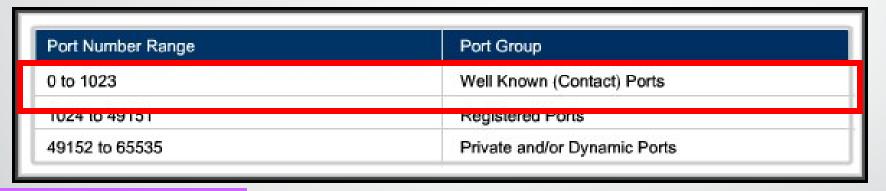
- Internet Corporation for Assigned Names and Numbers (ICANN) assigns port numbers.
- Three categories:



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



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.

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

8008 – Alternate HTTP

1863 – MSN Messenger

8080 – Alternate HTTP

5004 - RTP

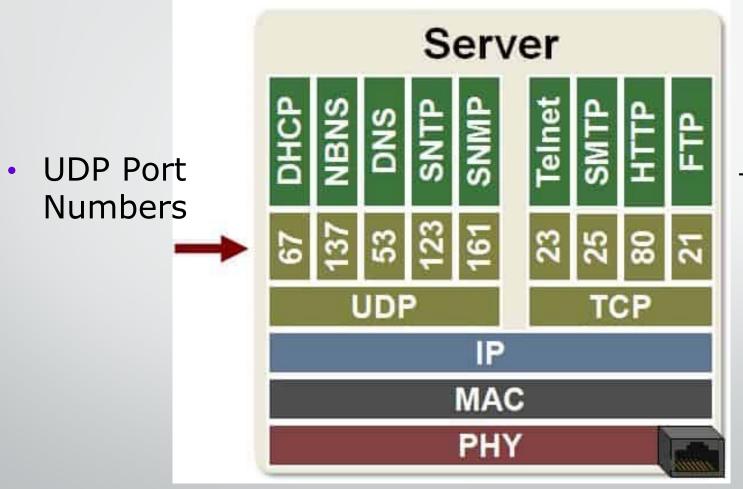
5060 - SIP (VoIP)

# 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**

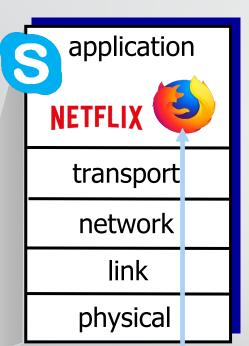


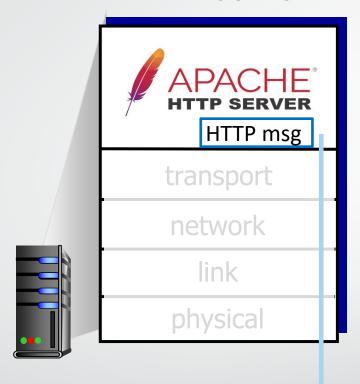
**TCP Port Numbers** 

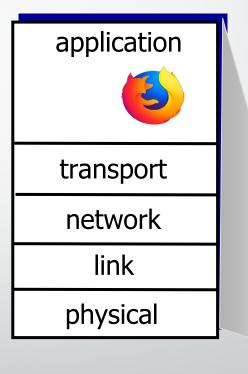
# Function 3 - Multiplexing

**HTTP** server





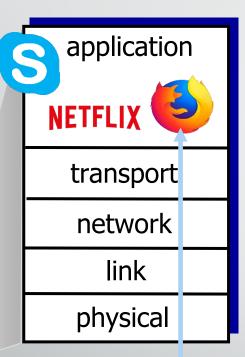


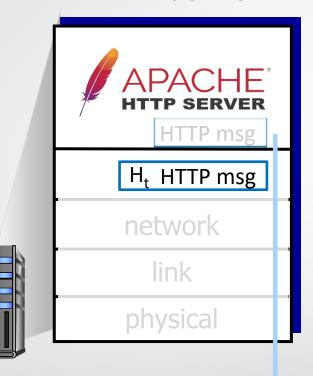


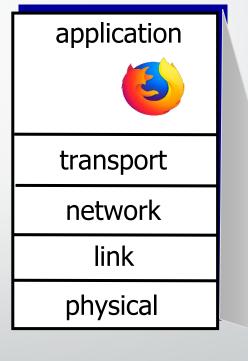


**HTTP** server



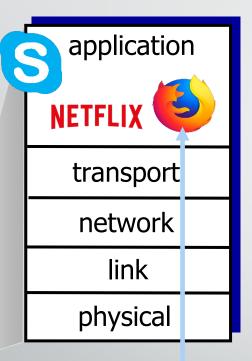


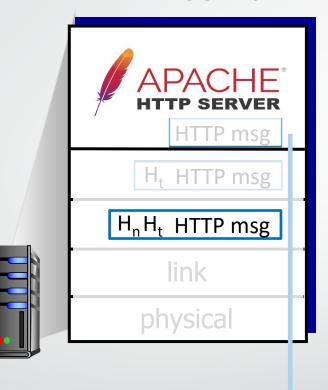


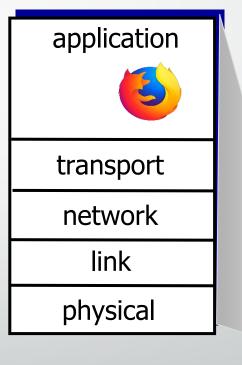


**HTTP** server





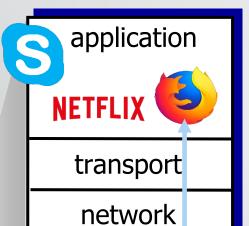






**HTTP** server

client



link physical



H<sub>n</sub>H<sub>t</sub> HTTP msg

application
transport
network
link
physical





### client<sub>1</sub>

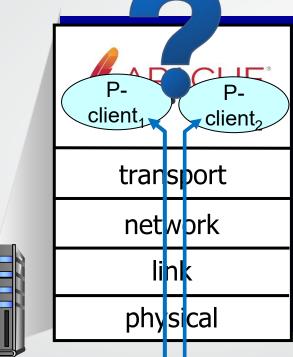


transport

network

link

physical



### client<sub>2</sub>





transport

network

link

physical



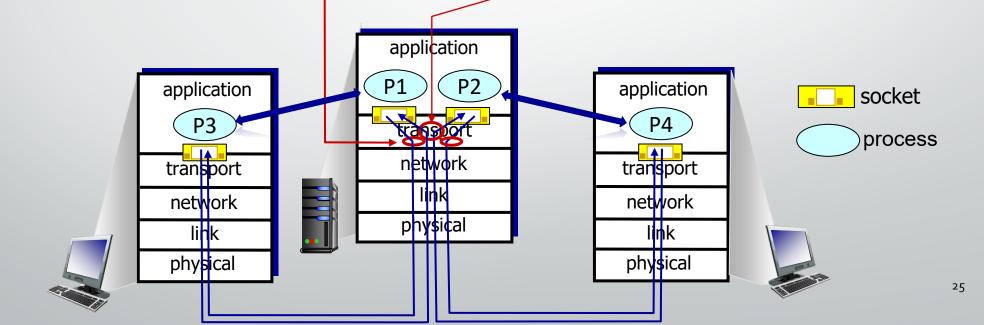
# Multiplexing/demultiplexing

### multiplexing at sender:

handle data from multiple sockets, add transport header (later used for demultiplexing)

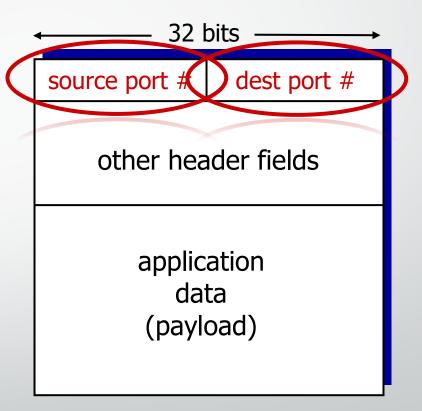
#### demultiplexing at receiver:

use header info to deliver received segments to correct socket



# 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 #

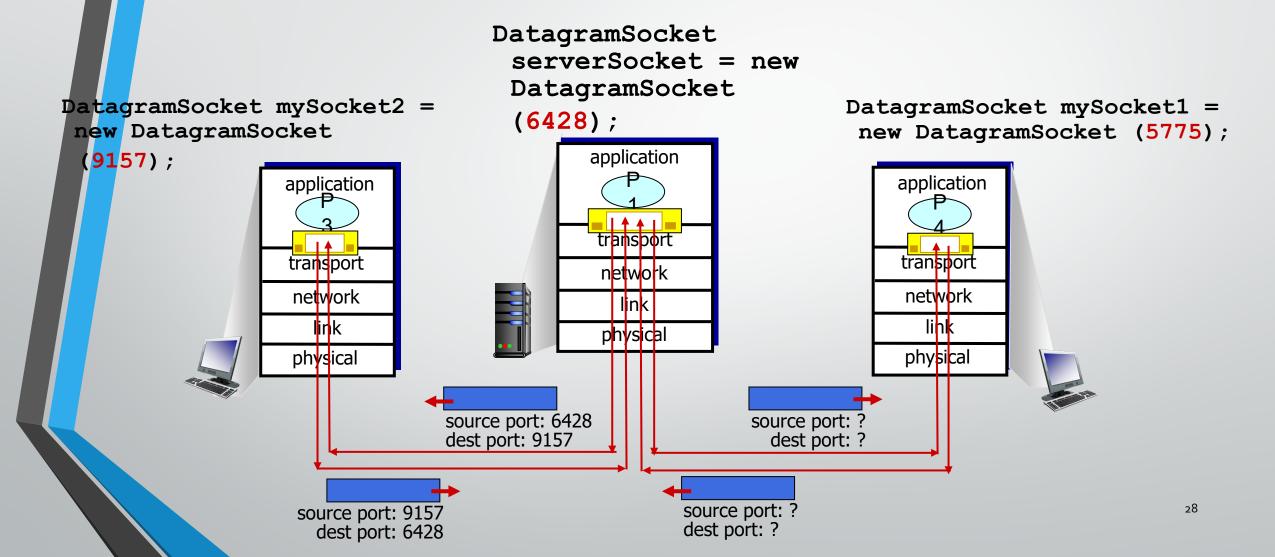
when receiving host receives *UDP* segment:

- 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

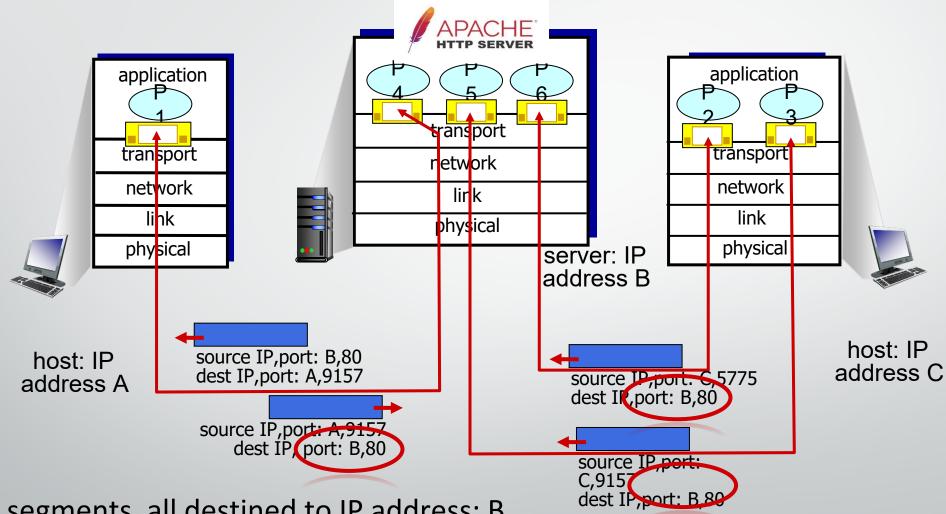


# Connection-oriented demultiplexing

- 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



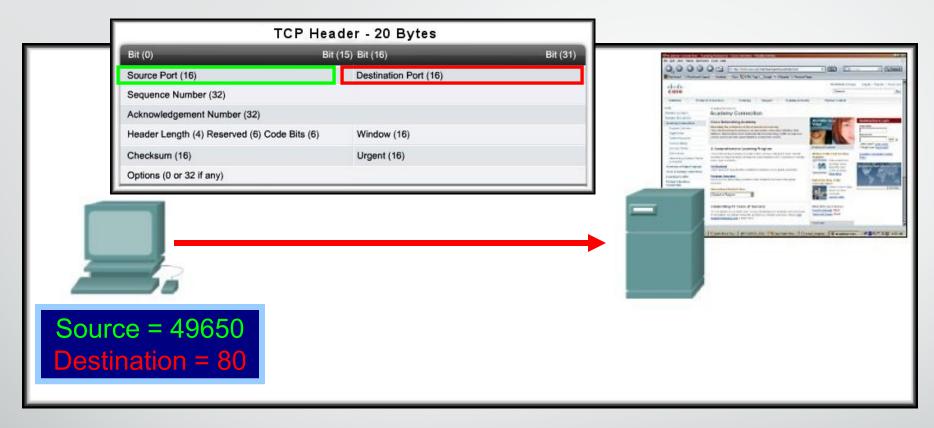
three segments, all destined to IP address: B,

dest port: 80 are demultiplexed to different sockets

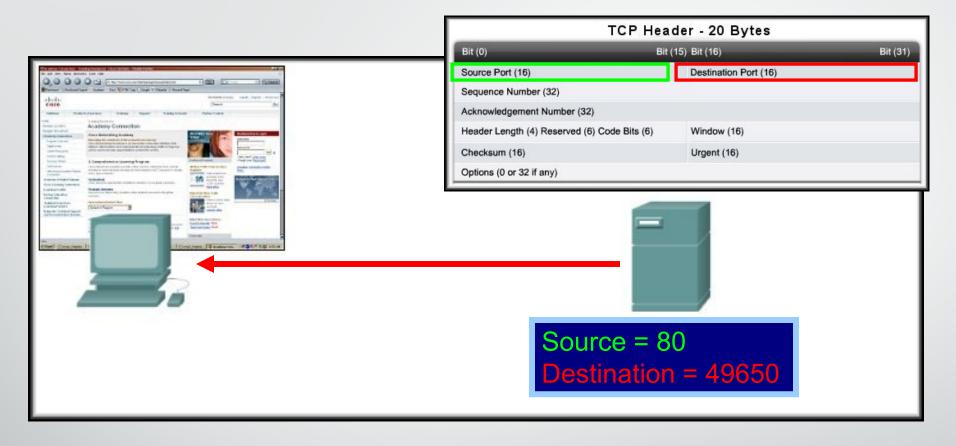
# Summary

- 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

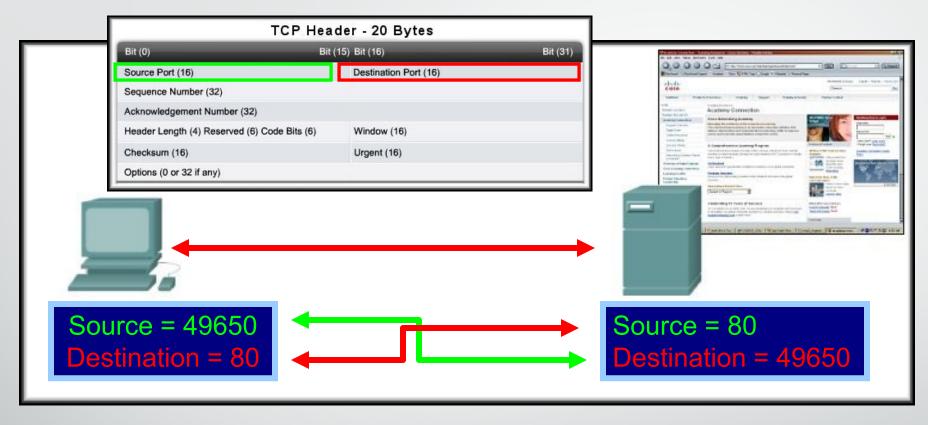
### More on Port Numbers



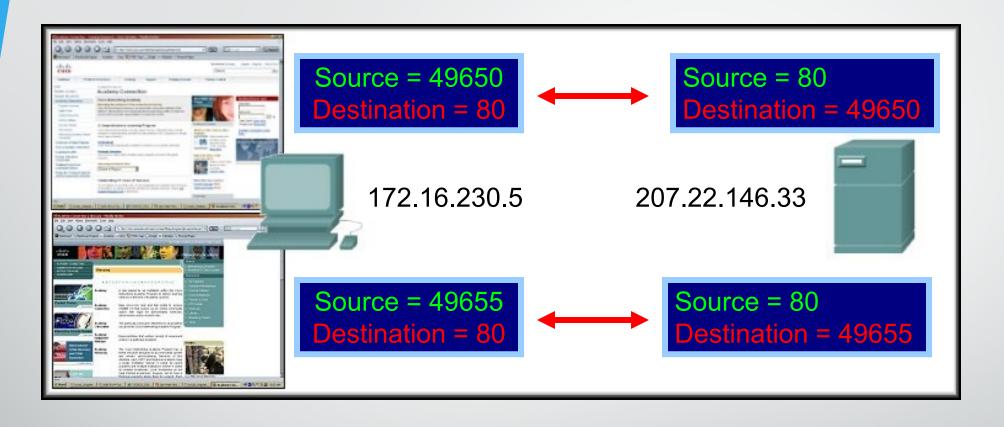
- Server is listening on Port 8o for HTTP connections.
- The client sets the destination port to 80 and uses a dynamic port as its source.



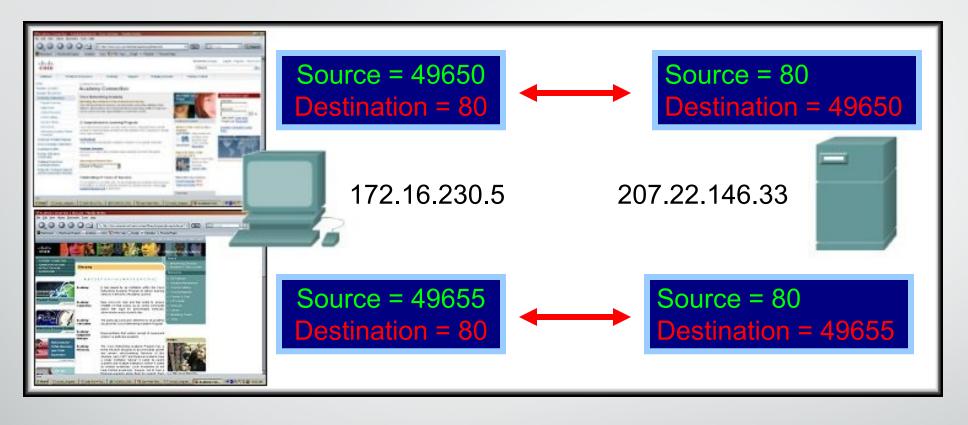
- Server replies with the web page.
  - Sets the source port to 80 and uses the client's source port as the destination.



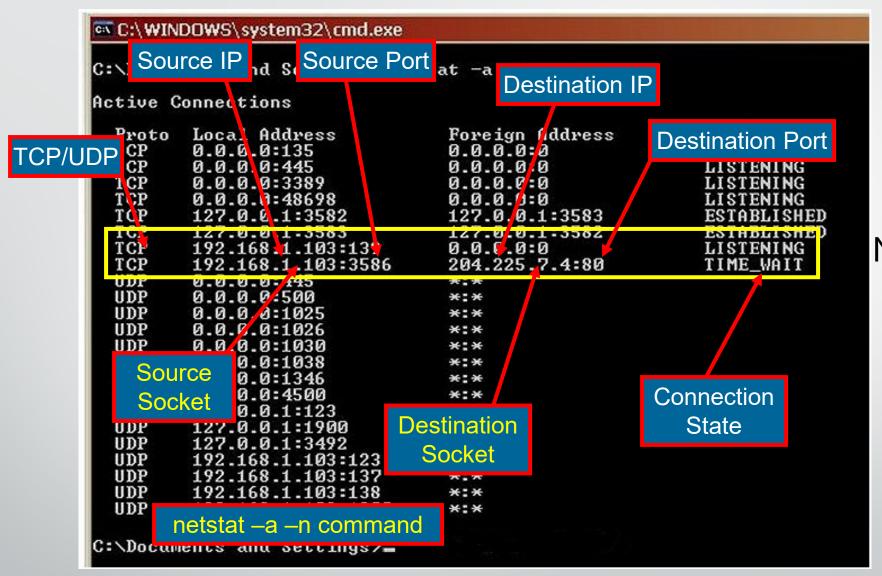
- Notice how the source and destination ports are used.
- Clients can use any random port number, servers can't.
  - Because clients won't be able to identify server process otherwise
- Servers, however, cannot use any random port number
  - Use of well-known port numbers!



- 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)



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 no 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)

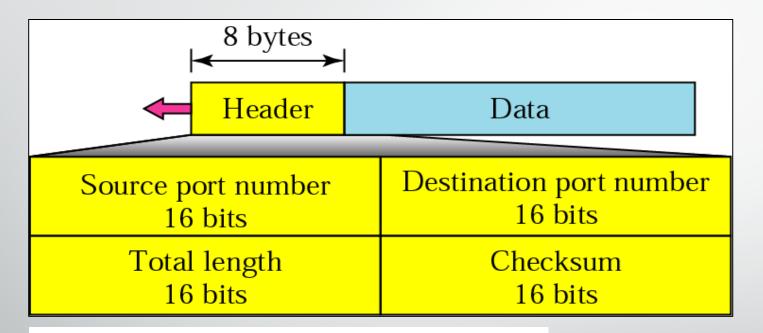
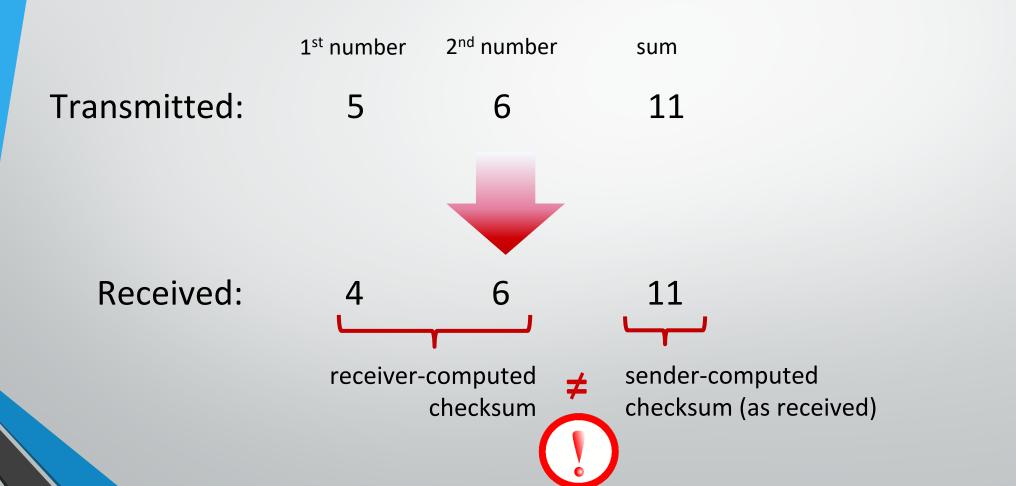


Figure 14: User Datagram Header format

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

# **UDP** checksum

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



### **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 into UDP 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

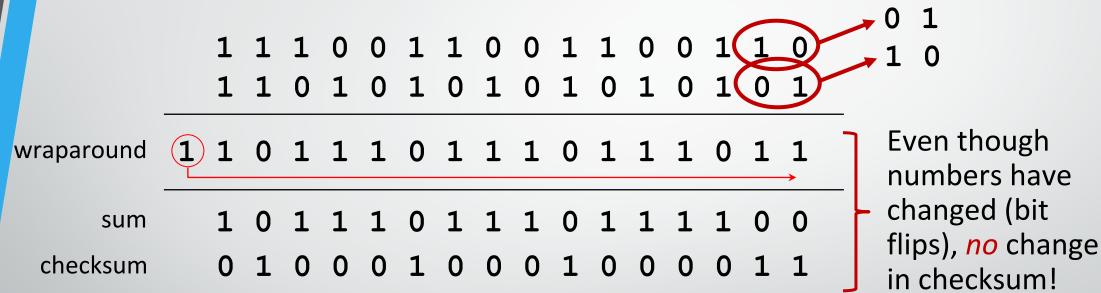
example: add two 16-bit integers



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