

Chapter 3: Transport layer

- How can two entities reliably communicate over a channel in which messages can be corrupted or lost?
- How can two distributed entities synchronize and share state?
- How can a collection of different entities adjust their communication rates to prevent network congestion and resource exhaustion?
- The two main transport protocols UDP and TCP.

Chapter 3: Transport layer: overview

Our goal:

- understand principles behind transport layer services:
 - multiplexing, demultiplexing
 - reliable data transfer
 - flow control
 - congestion control
- learn about Internet transport layer protocols:
 - UDP: connectionless transport
 - TCP: connection-oriented reliable transport
 - TCP congestion control

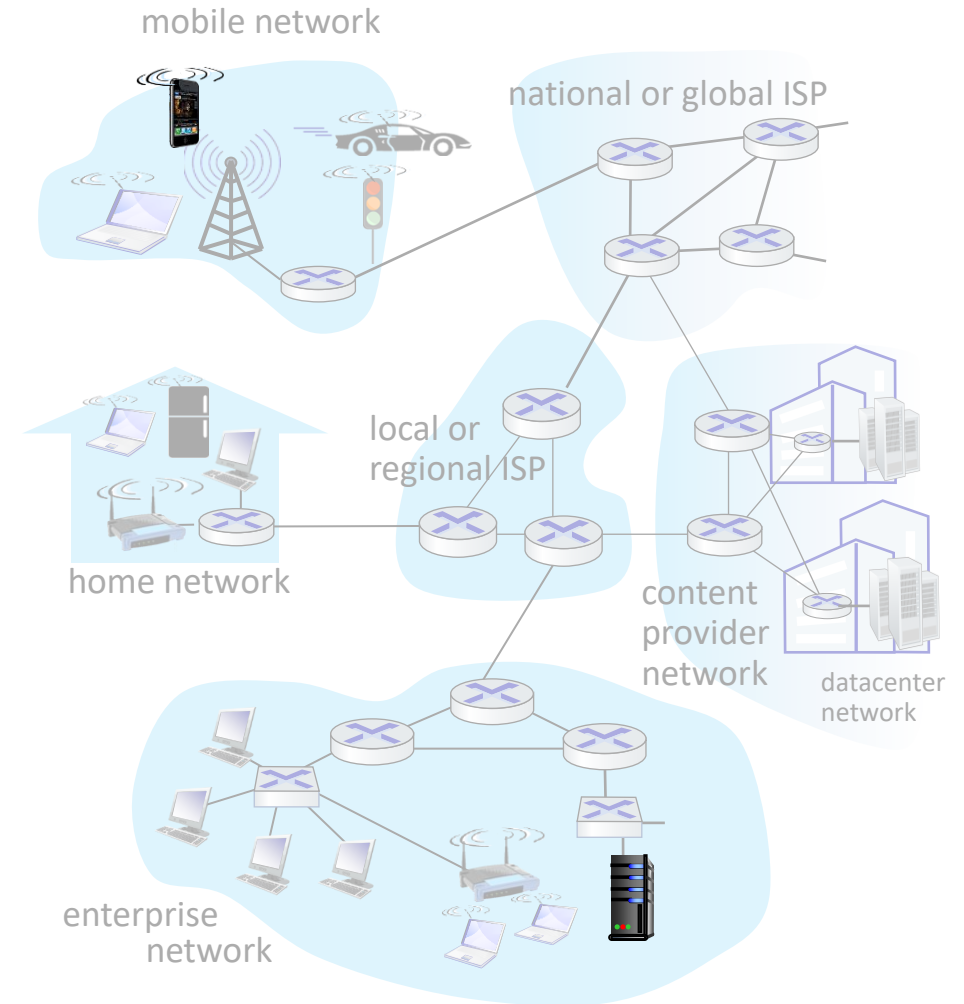
Transport layer: roadmap

- Transport-layer services
- Multiplexing and demultiplexing
- Connectionless transport: UDP
- Principles of reliable data transfer
- Connection-oriented transport: TCP
- Principles of congestion control
- TCP congestion control
- Evolution of transport-layer functionality



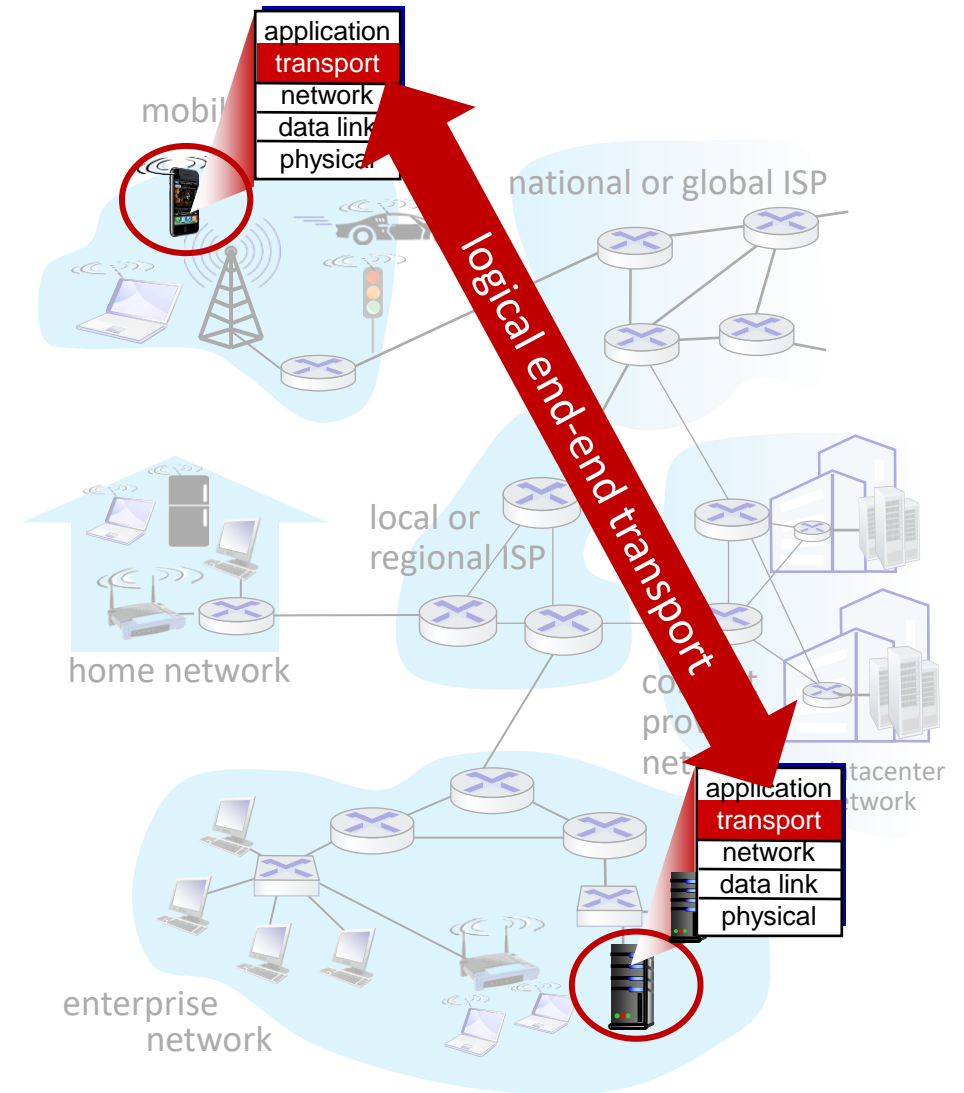
Transport services and protocols

- 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 services and protocols

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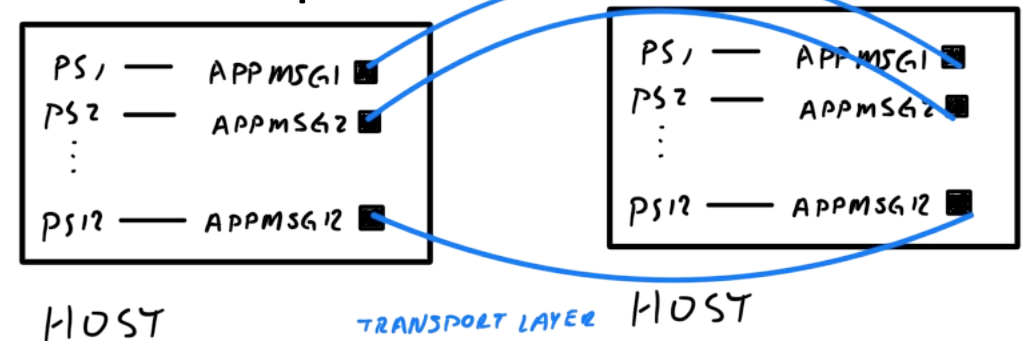
Transport vs. network layer services and protocols



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 vs. network layer services and protocols

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

household analogy:

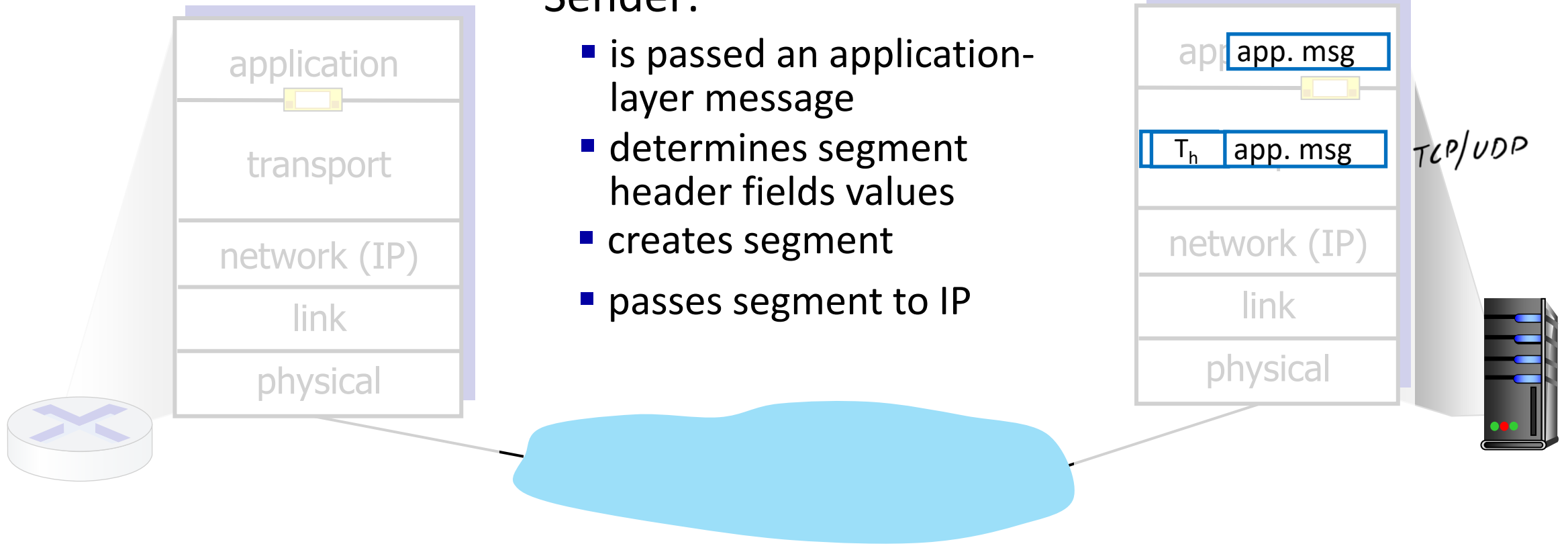
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Transport Layer Actions

Sender:

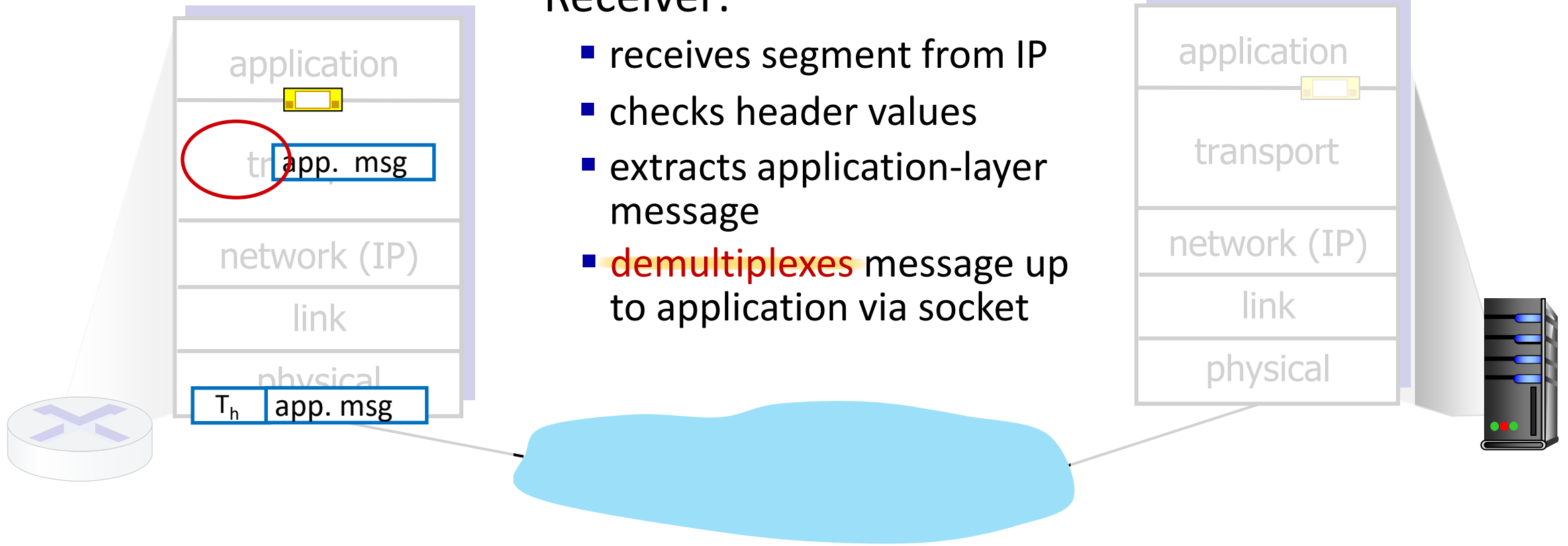
- is passed an application-layer message
- determines segment header fields values
- creates segment
- passes segment to IP



Transport Layer Actions

Receiver:

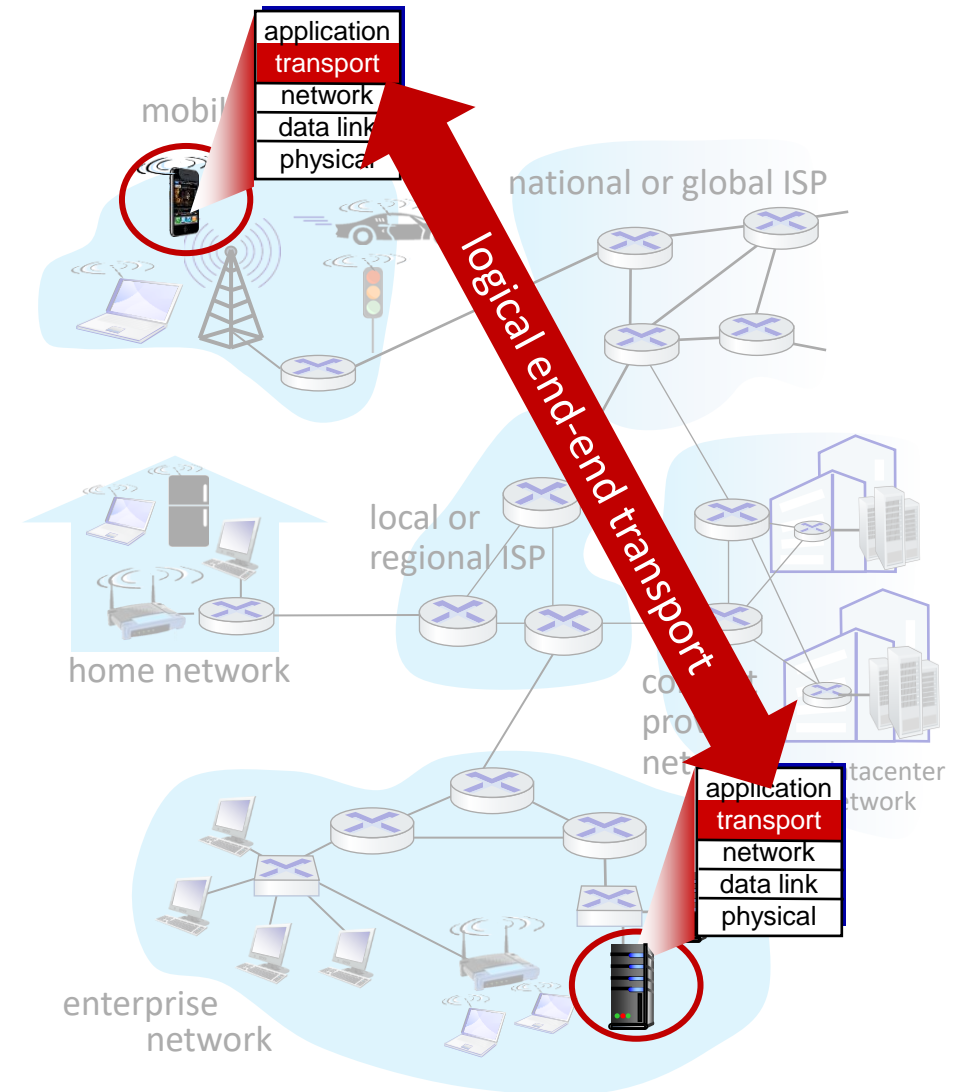
- receives segment from IP
- checks header values
- extracts application-layer message
- **demultiplexes** message up to application via socket



Two principal Internet transport protocols

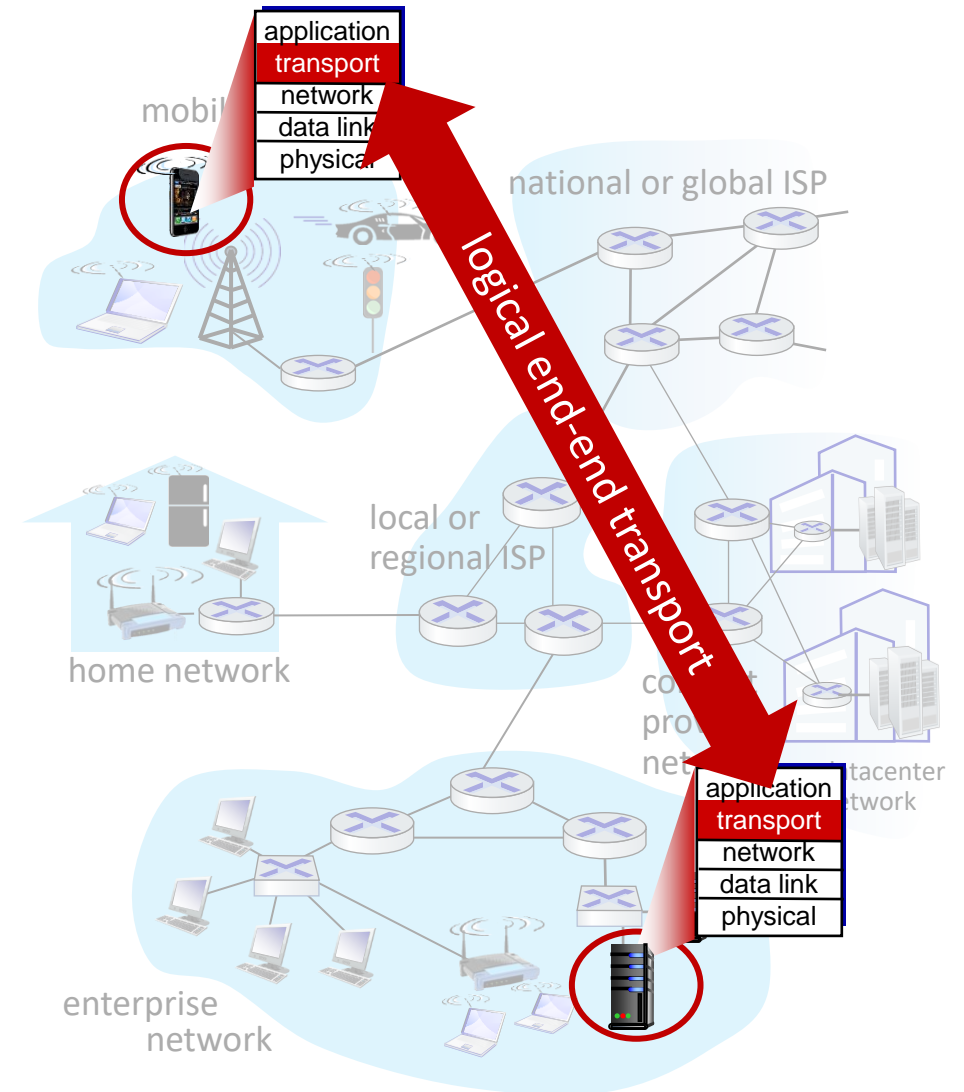
■ **TCP:** Transmission Control Protocol

- reliable, in-order delivery
- congestion control
- flow control
- connection setup



Two principal Internet transport protocols

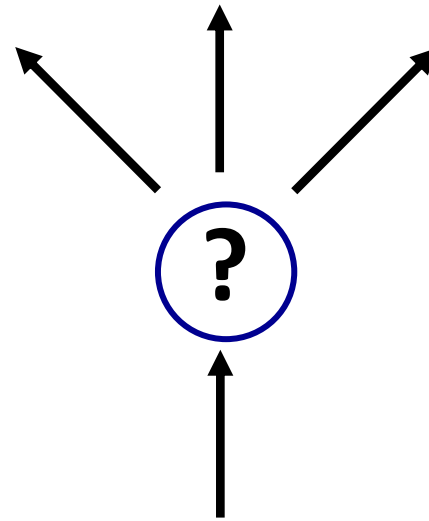
- **UDP:** User Datagram Protocol
 - unreliable, unordered delivery
 - no-frills extension of “best-effort” IP
- services *not* available:
 - delay guarantees
 - bandwidth guarantees



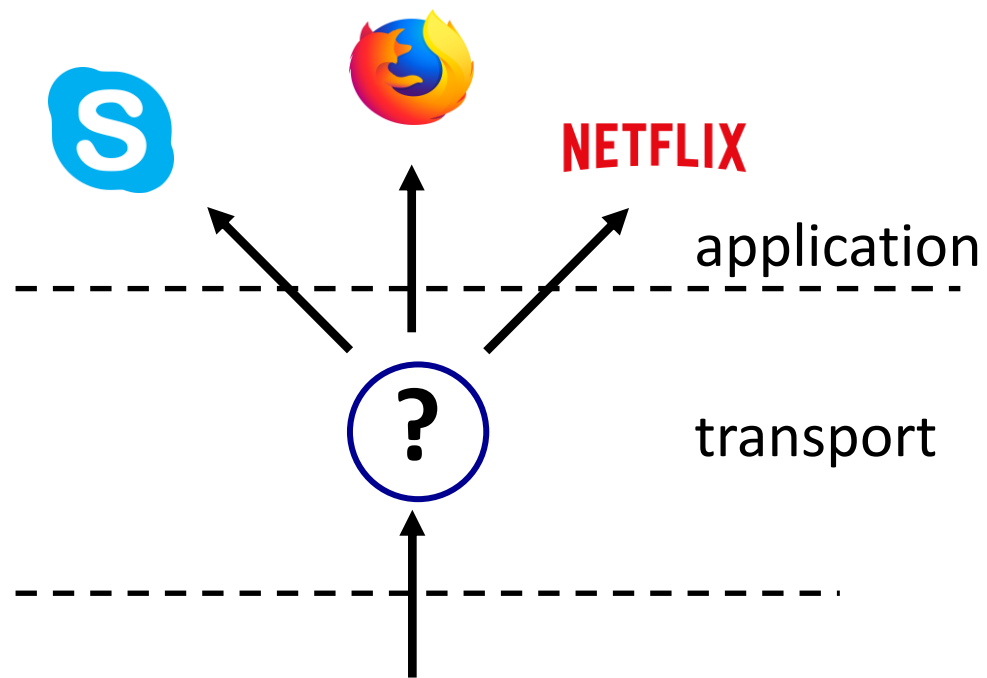
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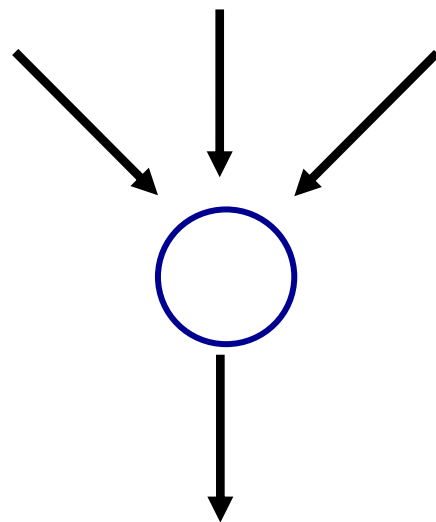




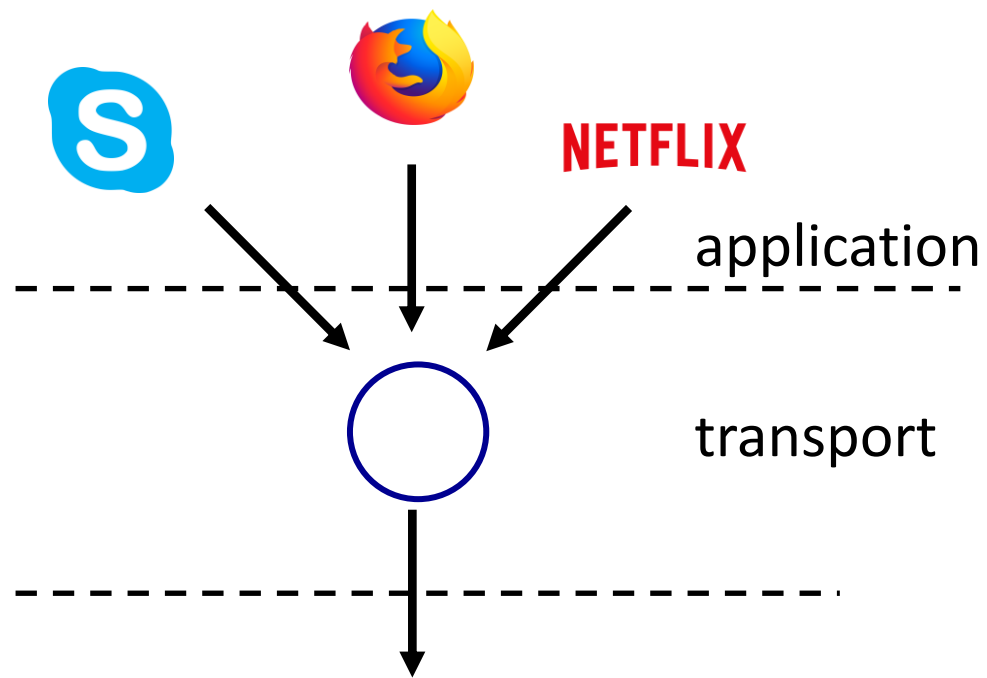
de-multiplexing



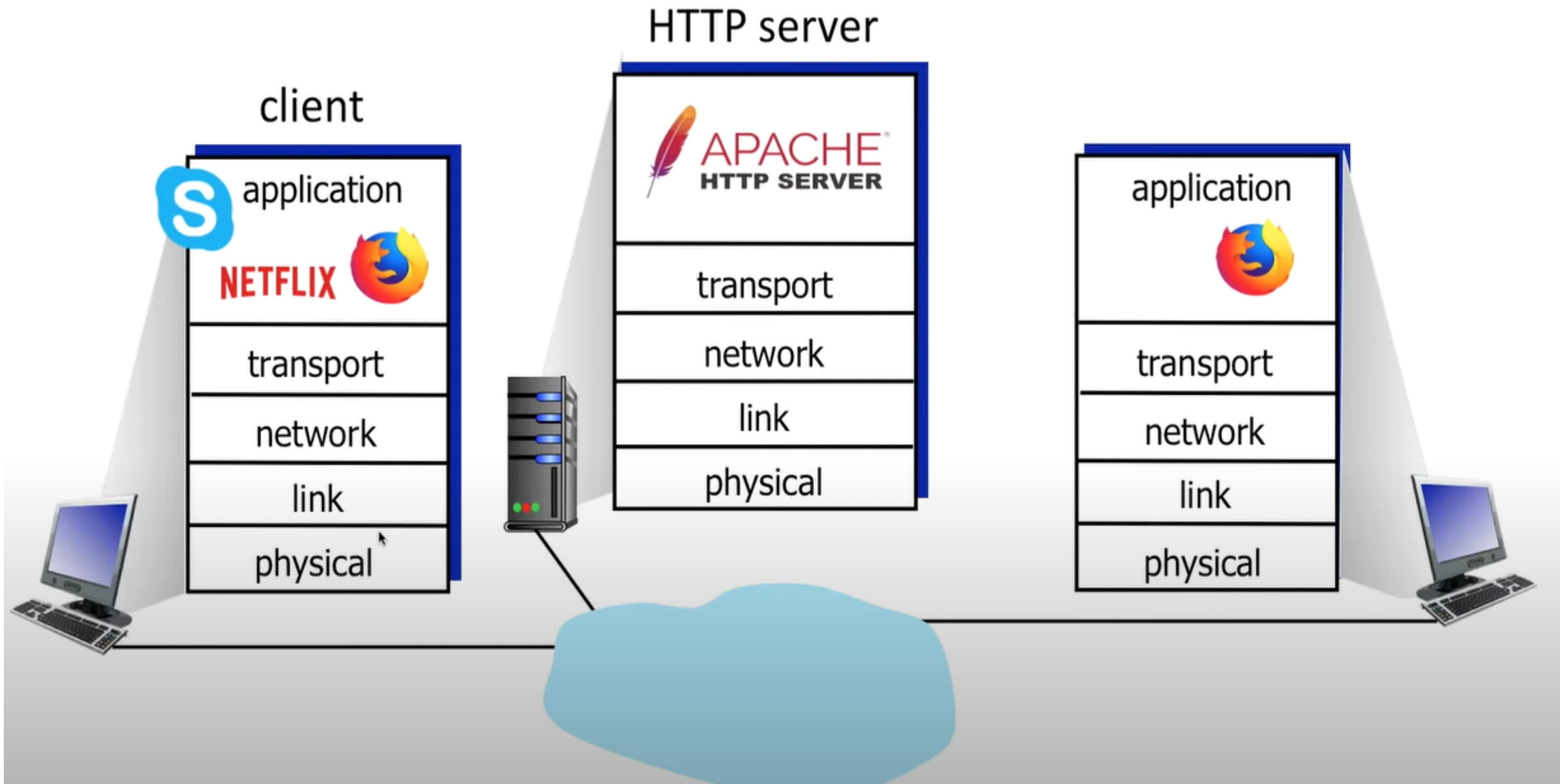
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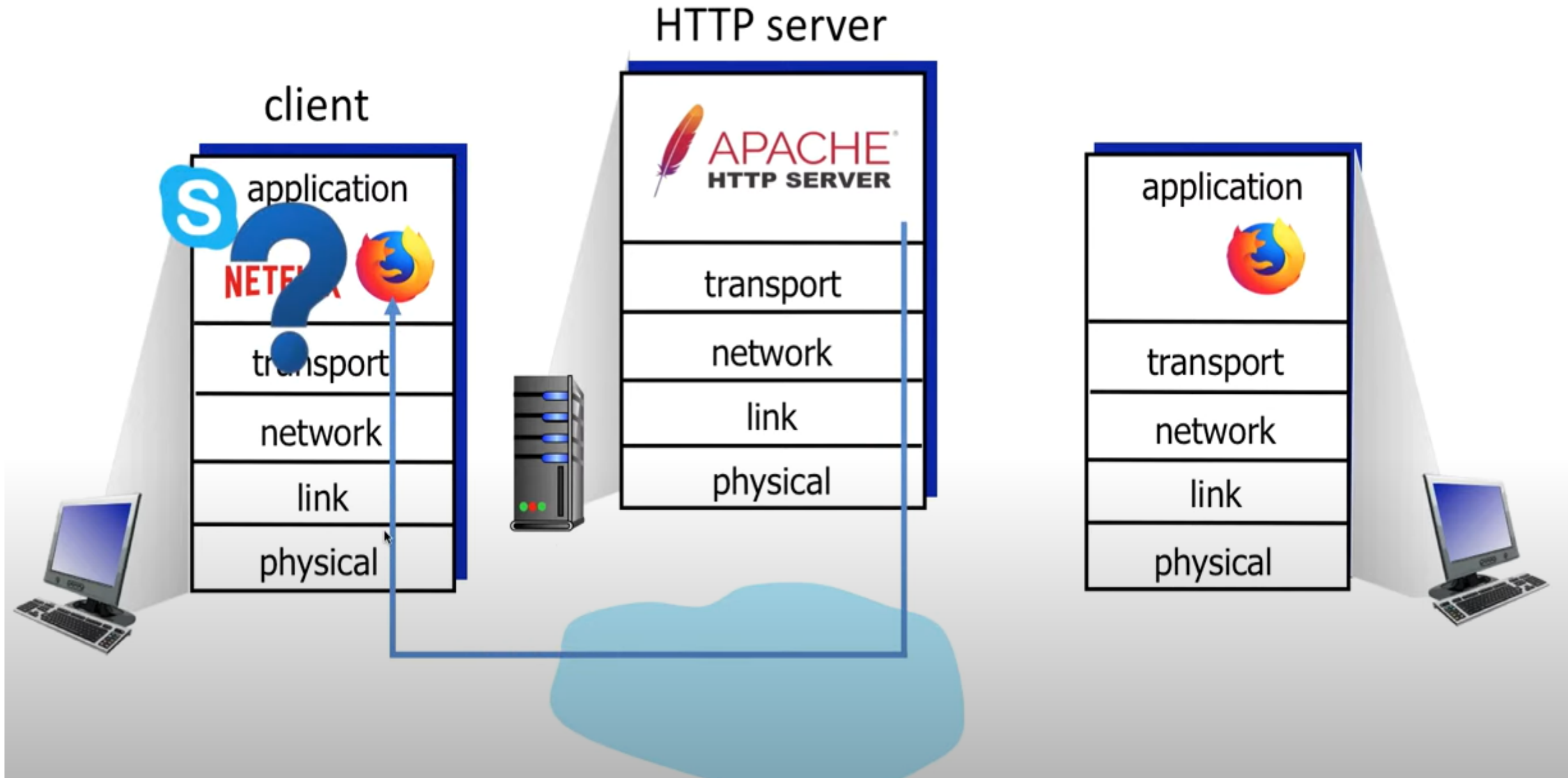


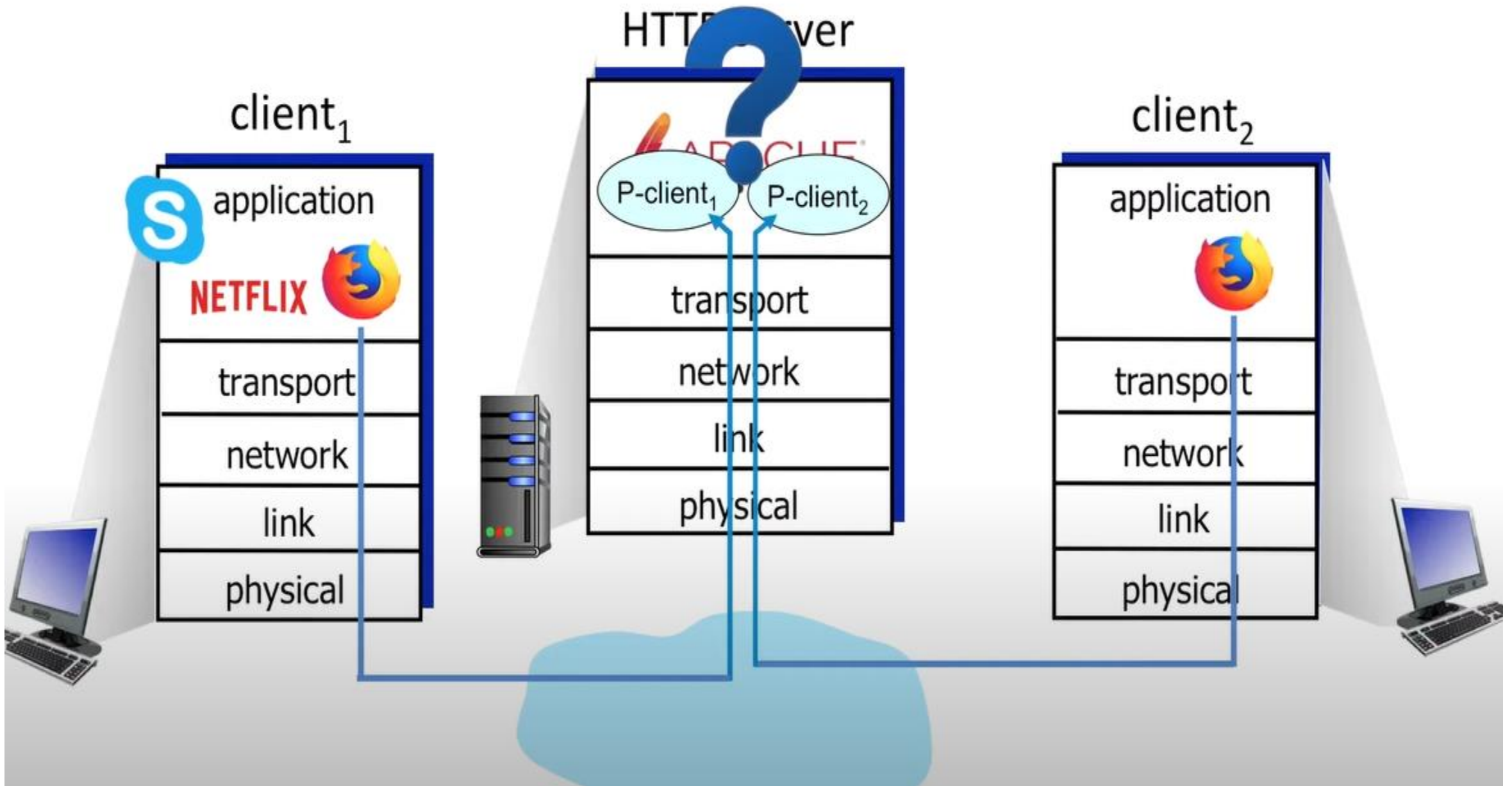
multiplexing



multiplexing









Multiplexing



Demultiplexing

AIRFRANCE 

ECONOMY 



AIRFRANCE 

SKY
PRIORITY™



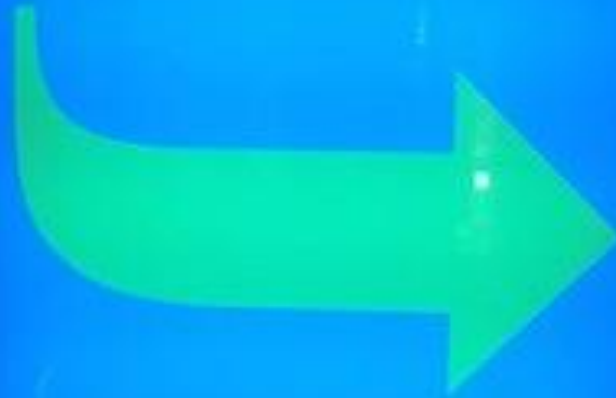
TSA Pre✓



Transportation
Security
Administration

tsa.gov

Main
Checkpoint



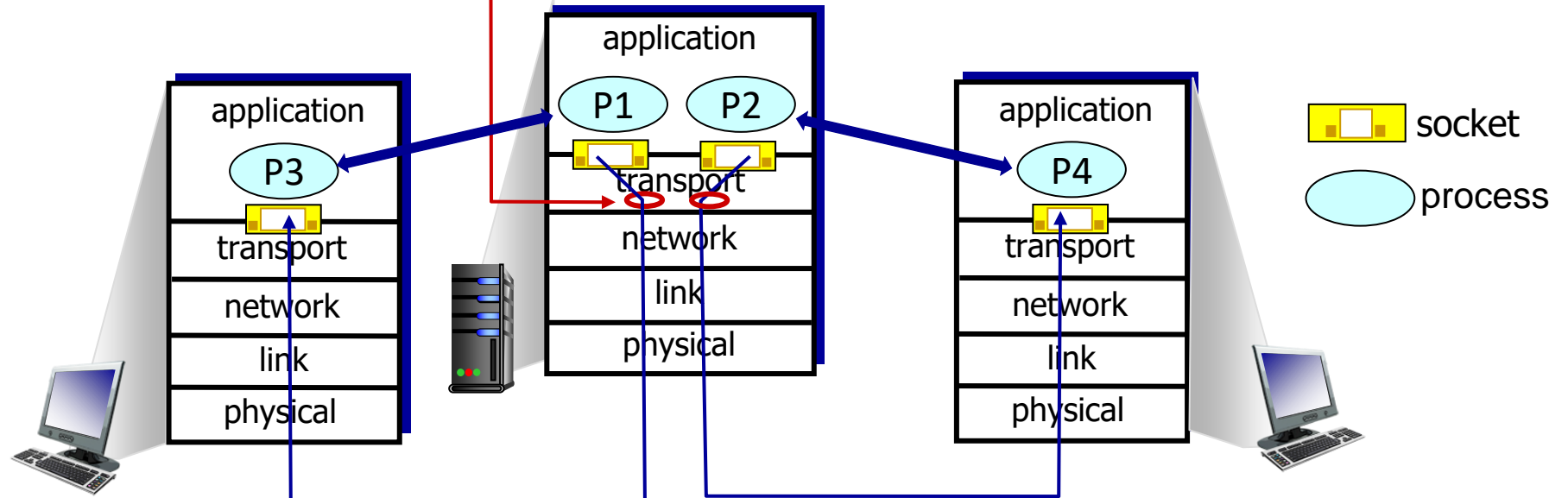
Transportation
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Multiplexing/demultiplexing

multiplexing as sender:

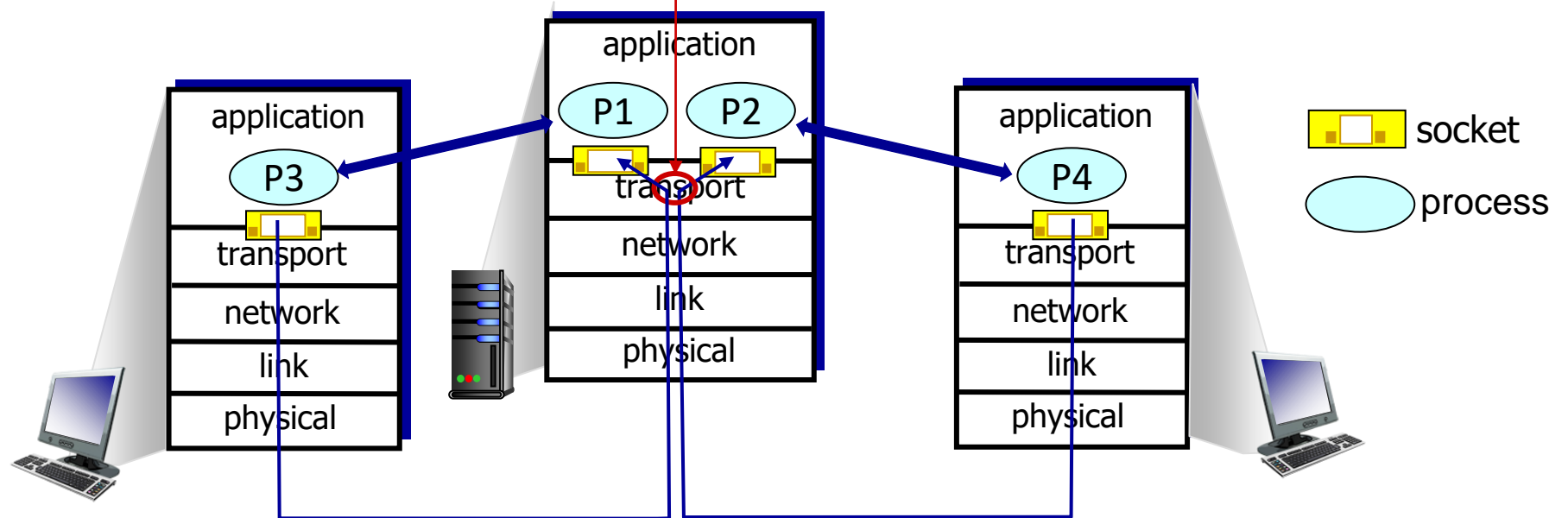
handle data from multiple sockets, add transport header (later used for demultiplexing)



Multiplexing/demultiplexing

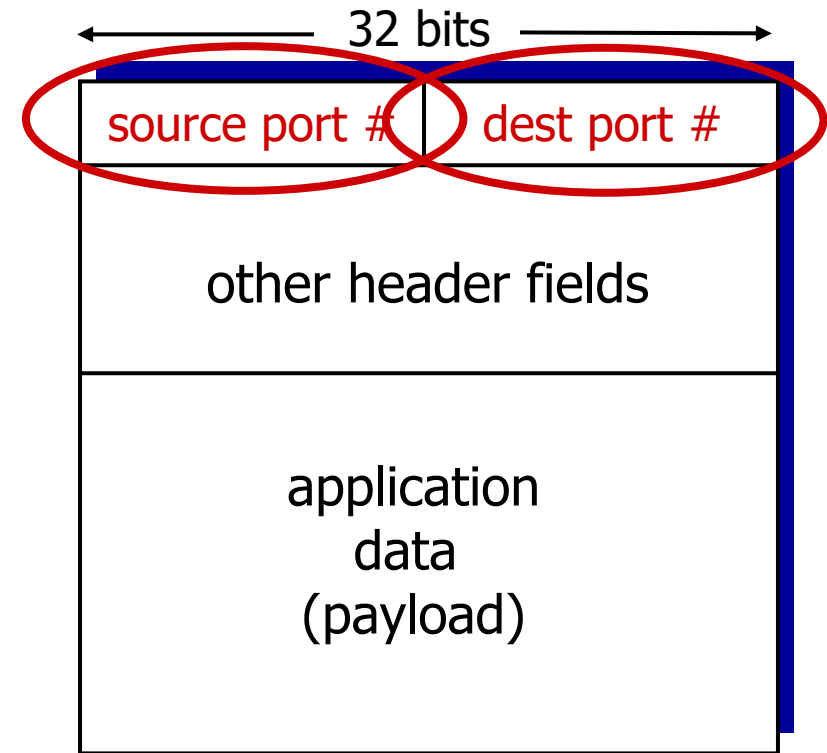
demultiplexing as receiver:

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

Recall:

- when creating socket, must specify *host-local* port #:

```
DatagramSocket mySocket1  
= new DatagramSocket(12534);
```

- 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 #



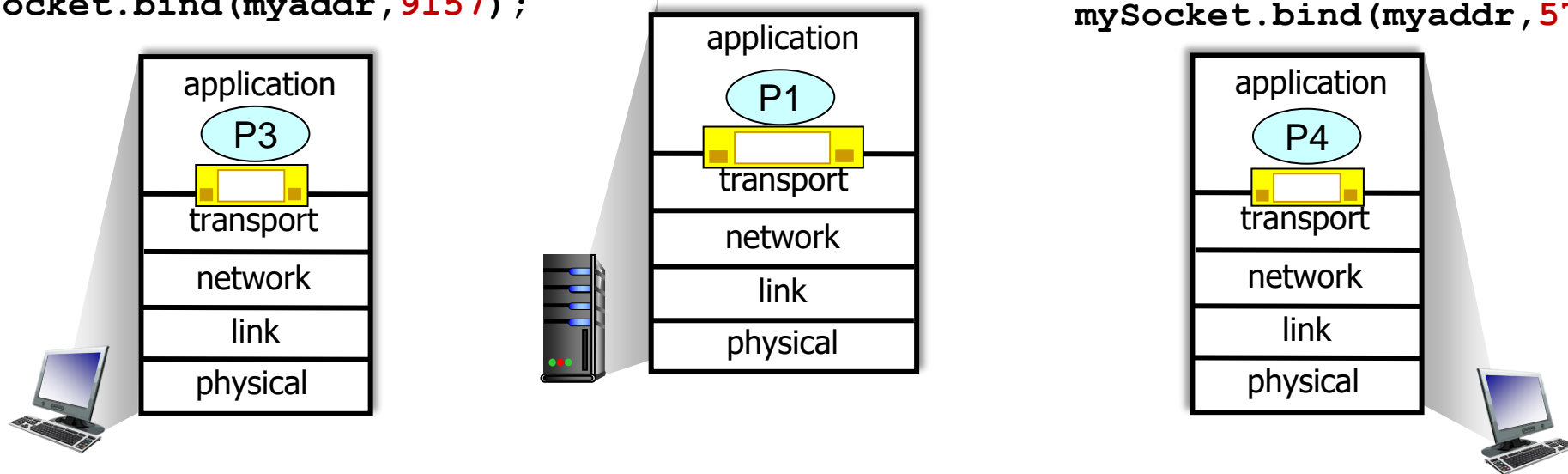
IP/UDP datagrams with *same dest. port #*, but different source IP addresses and/or source port numbers will be directed to *same socket* at receiving host

Connectionless demultiplexing: an example

```
mySocket =  
    socket(AF_INET, SOCK_DGRAM)  
mySocket.bind(myaddr, 6428);
```

```
mySocket =  
    socket(AF_INET, SOCK_STREAM)  
mySocket.bind(myaddr, 9157);
```

```
mySocket =  
    socket(AF_INET, SOCK_STREAM)  
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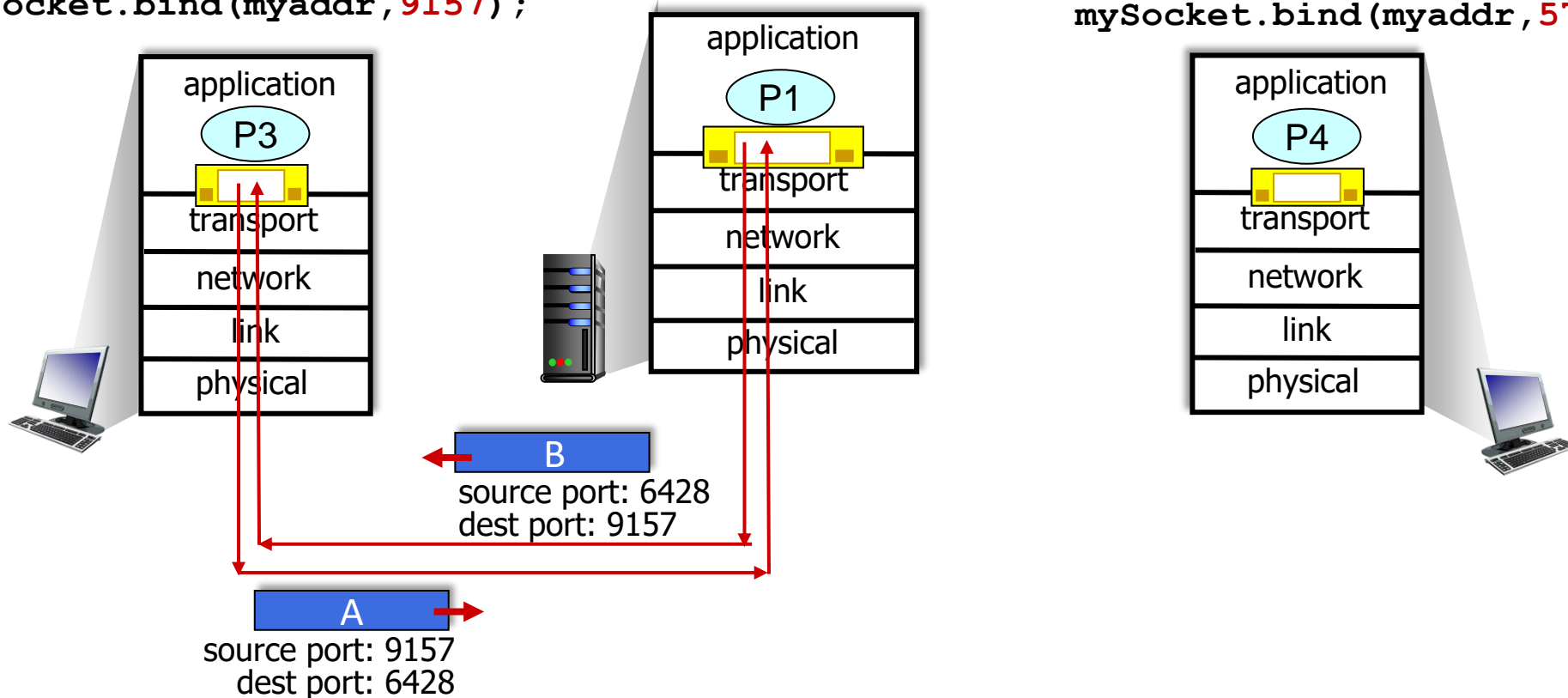


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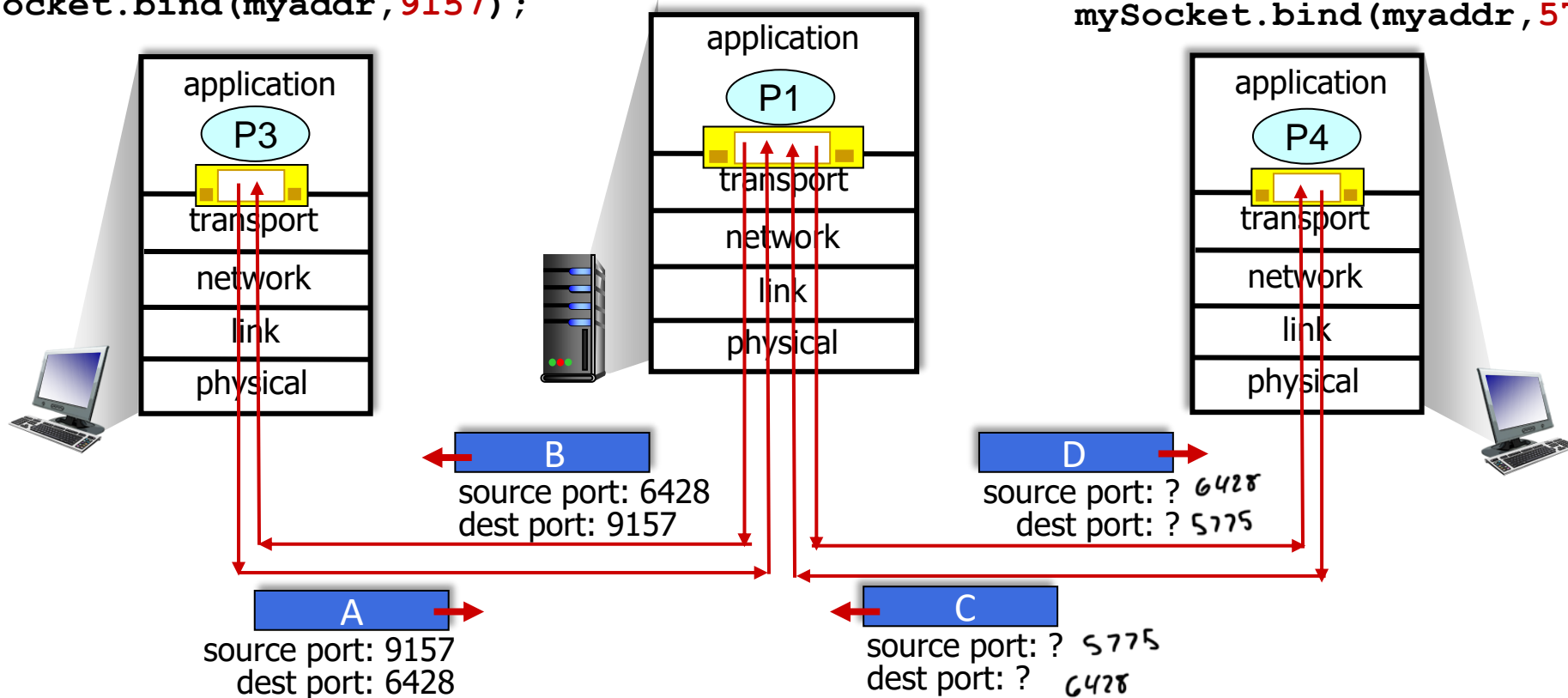


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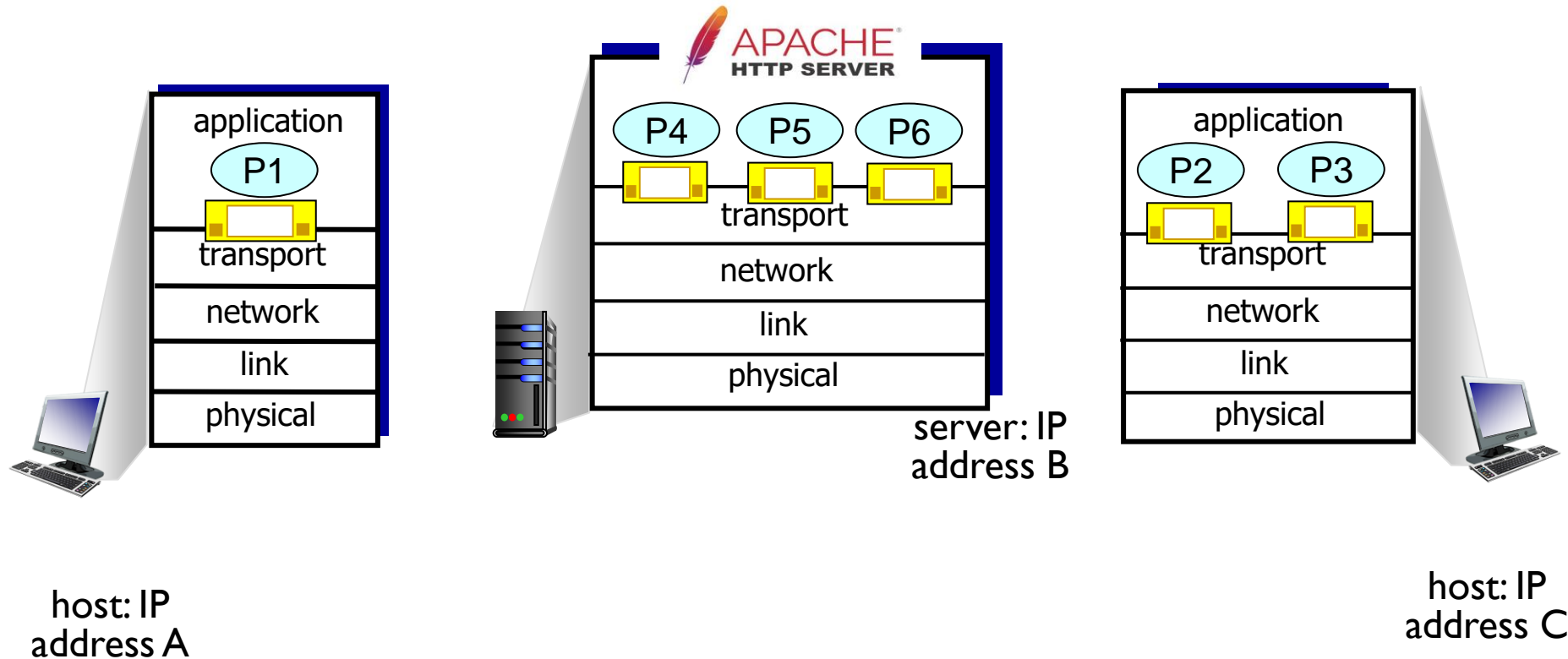
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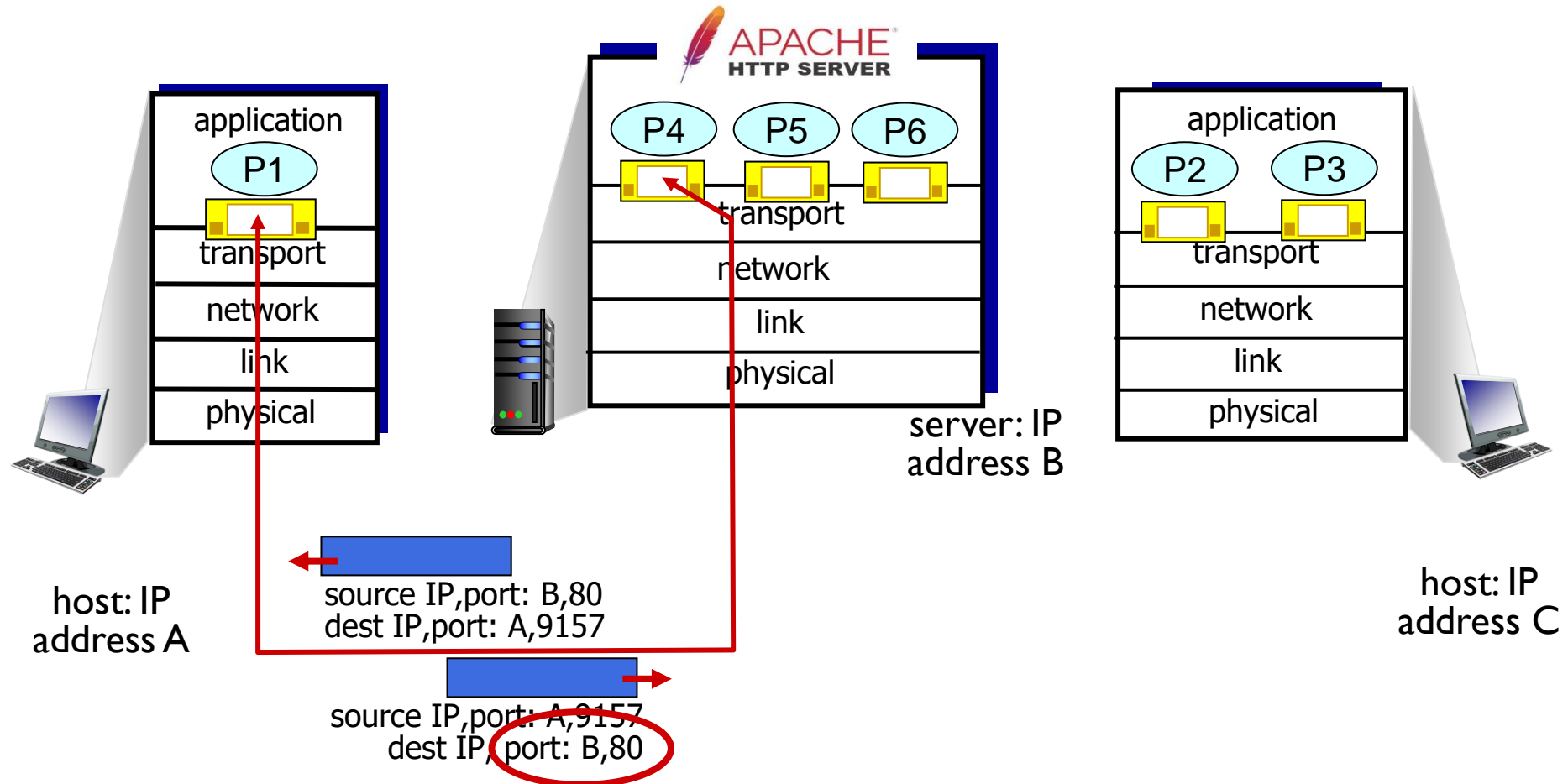
Connection-oriented demultiplexing

- 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

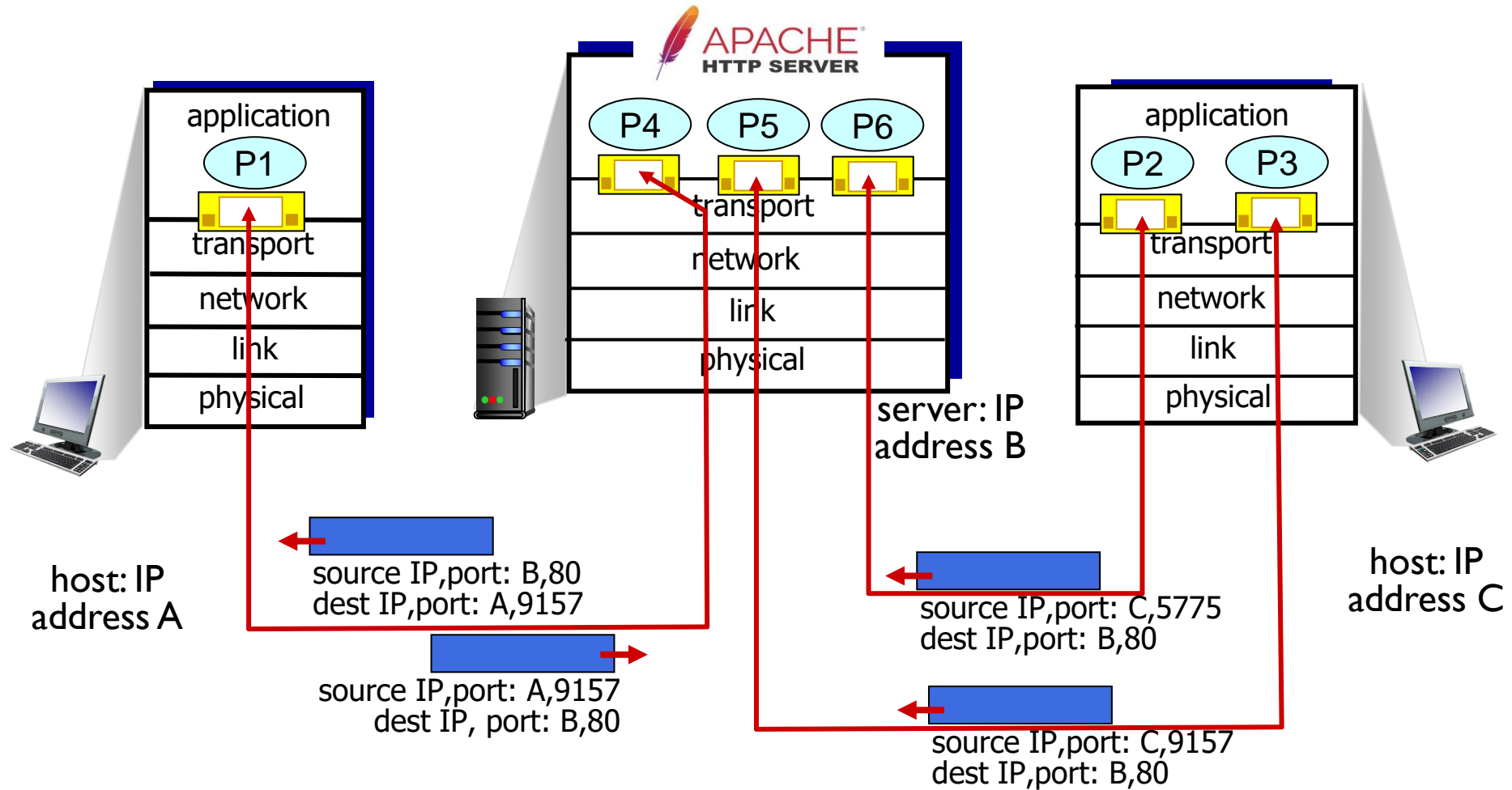
Connection-oriented demultiplexing: example



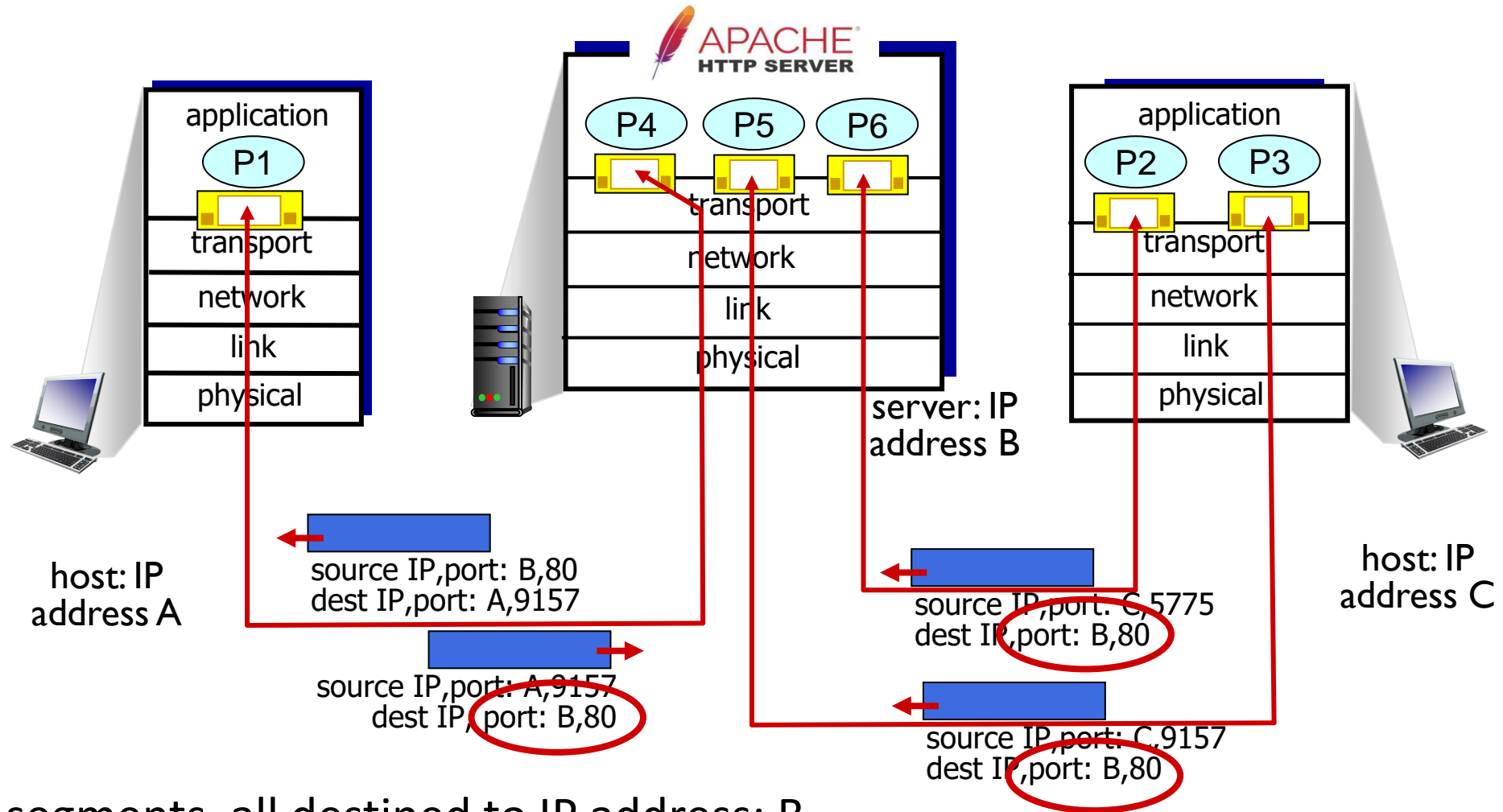
Connection-oriented demultiplexing: example



Connection-oriented demultiplexing: example



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

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- **Connectionless transport: UDP**
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UDP: User Datagram Protocol

- UDP sender/receiver actions
- UDP segment format
- Internet checksum

UDP: User Datagram Protocol

- “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

UDP: User Datagram Protocol

Why is there a UDP?

- no connection establishment (which can add RTT delay)
- simple: no connection state at sender, receiver
- small header size
- no congestion control
 - UDP can blast away as fast as desired!
 - can function in the face of congestion

UDP: User Datagram Protocol

- UDP use:
 - streaming multimedia apps (loss tolerant, rate sensitive)
 - DNS
 - SNMP
 - HTTP/3
- if reliable transfer needed over UDP (e.g., HTTP/3):
 - add needed reliability at application layer
 - add congestion control at application layer

UDP: User Datagram Protocol [RFC 768]

INTERNET STANDARD

RFC 768

J. Postel

ISI

28 August 1980

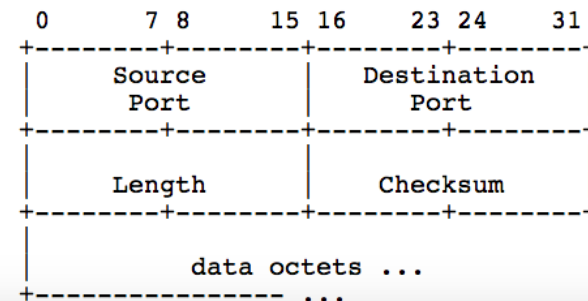
User Datagram Protocol

Introduction

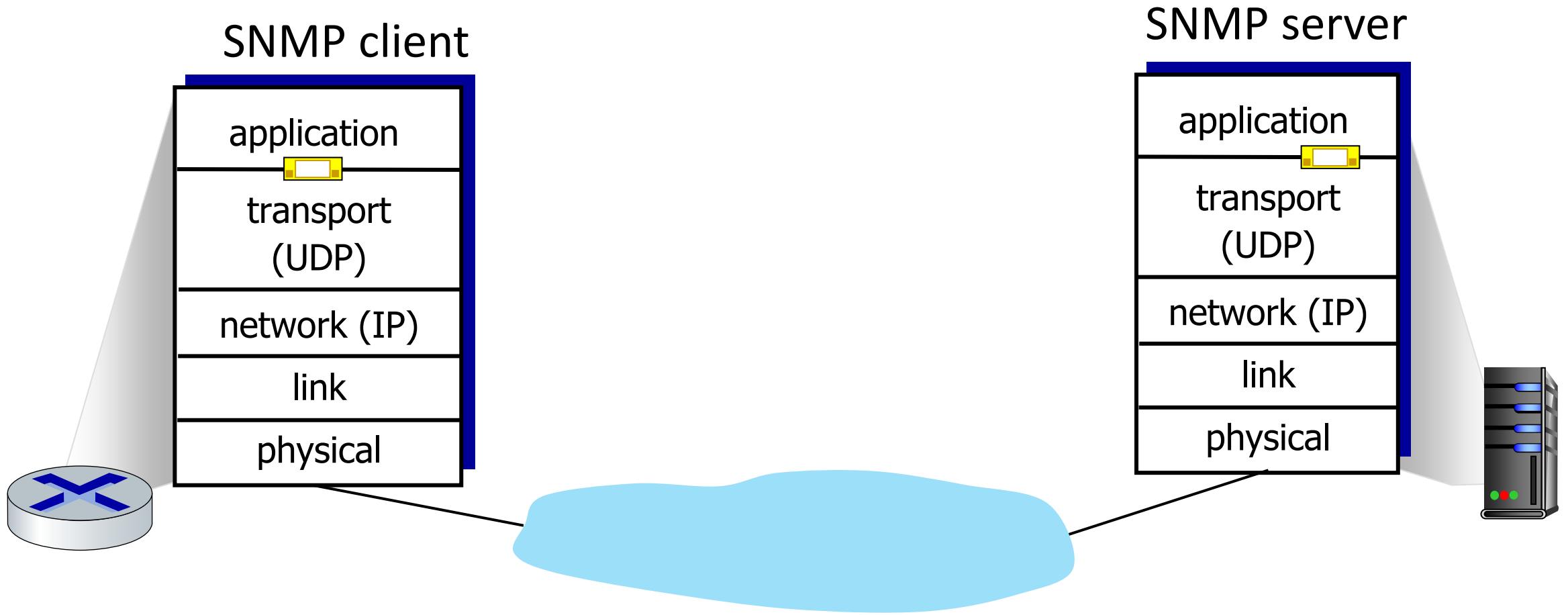
This User Datagram Protocol (UDP) is defined to make available a datagram mode of packet-switched computer communication in the environment of an interconnected set of computer networks. This protocol assumes that the Internet Protocol (IP) [1] is used as the underlying protocol.

This protocol provides a procedure for application programs to send messages to other programs with a minimum of protocol mechanism. The protocol is transaction oriented, and delivery and duplicate protection are not guaranteed. Applications requiring ordered reliable delivery of streams of data should use the Transmission Control Protocol (TCP) [2].

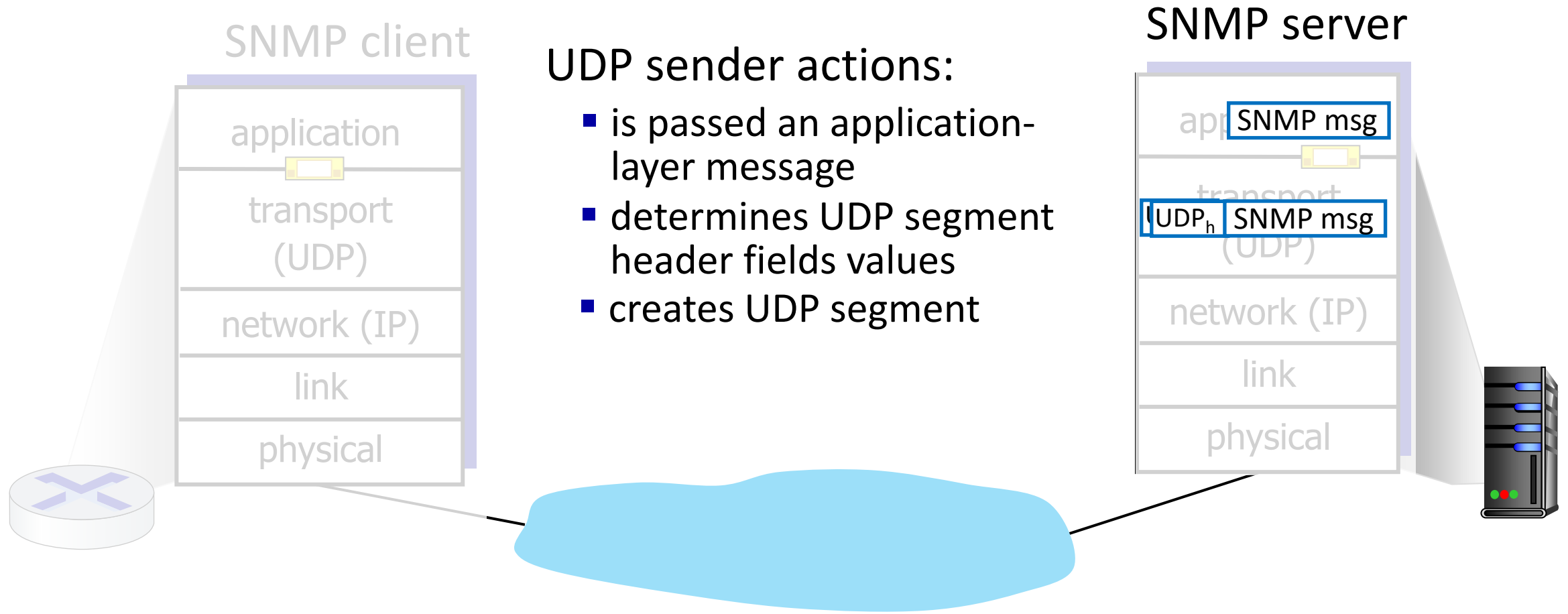
Format



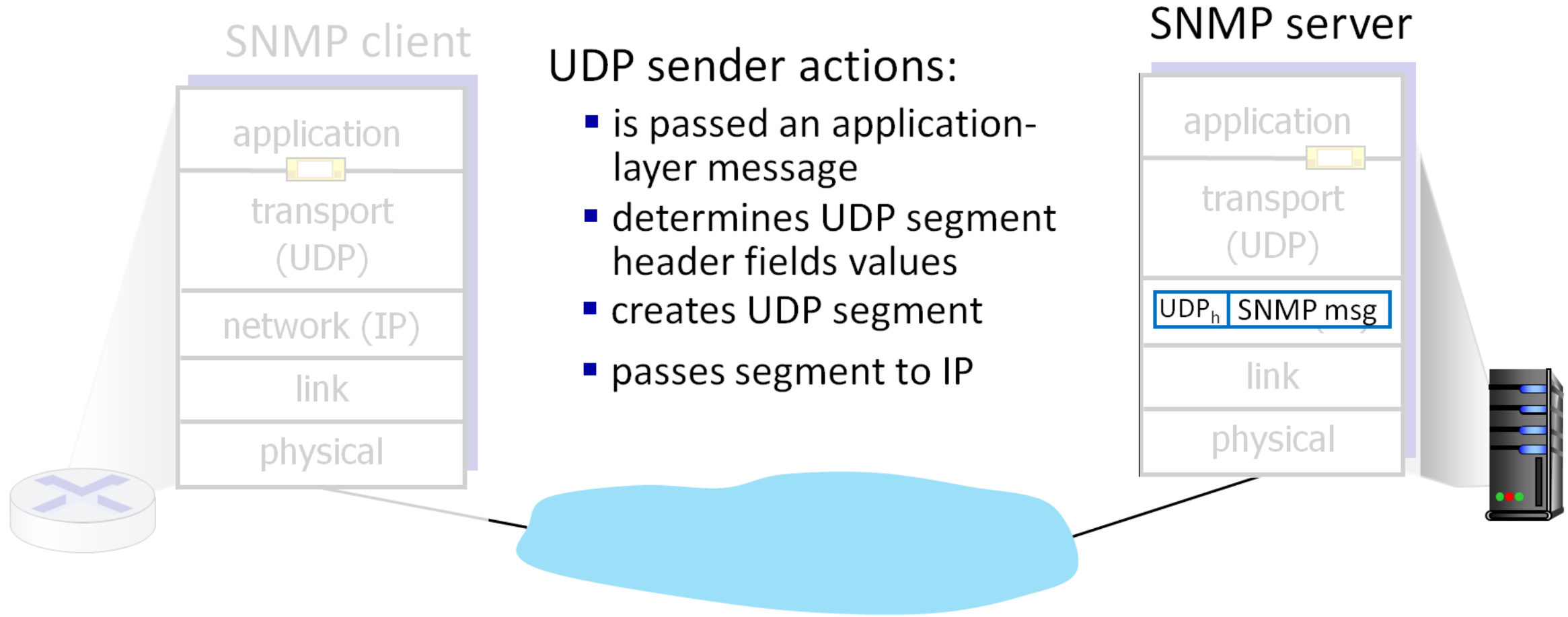
UDP: Transport Layer Actions



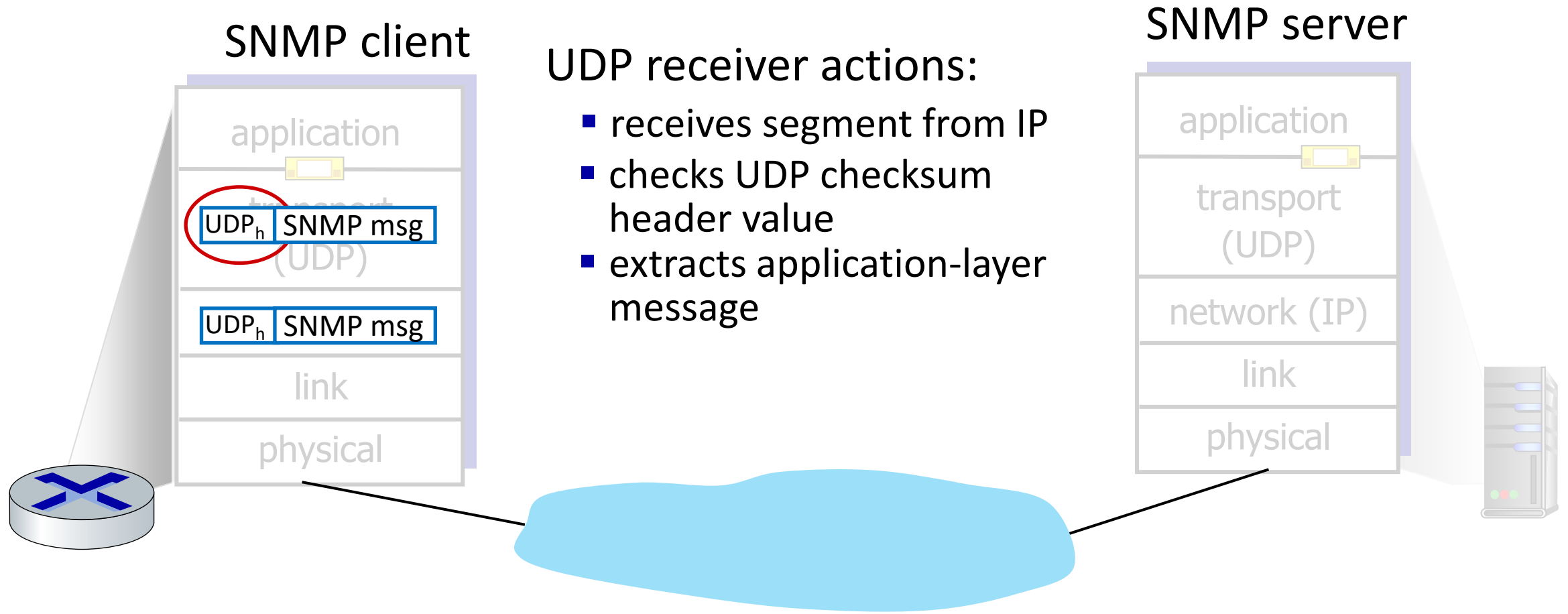
UDP: Transport Layer Actions



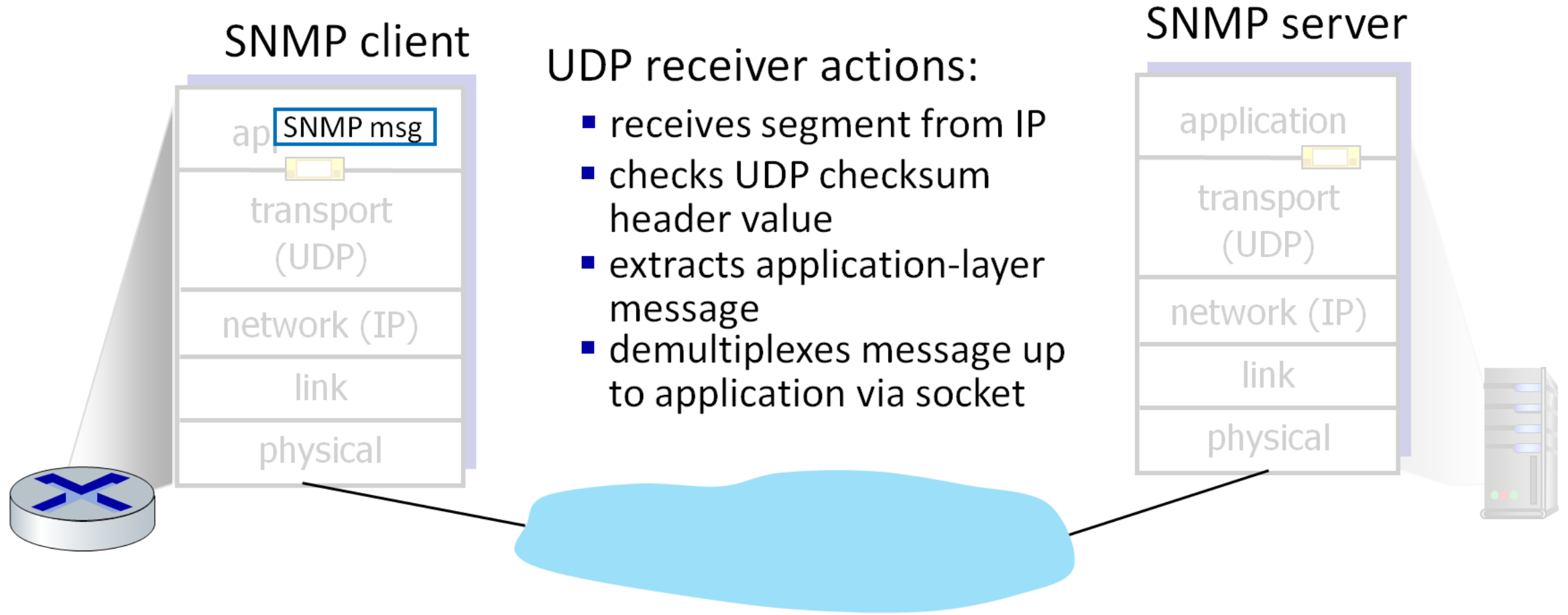
UDP: Transport Layer Actions



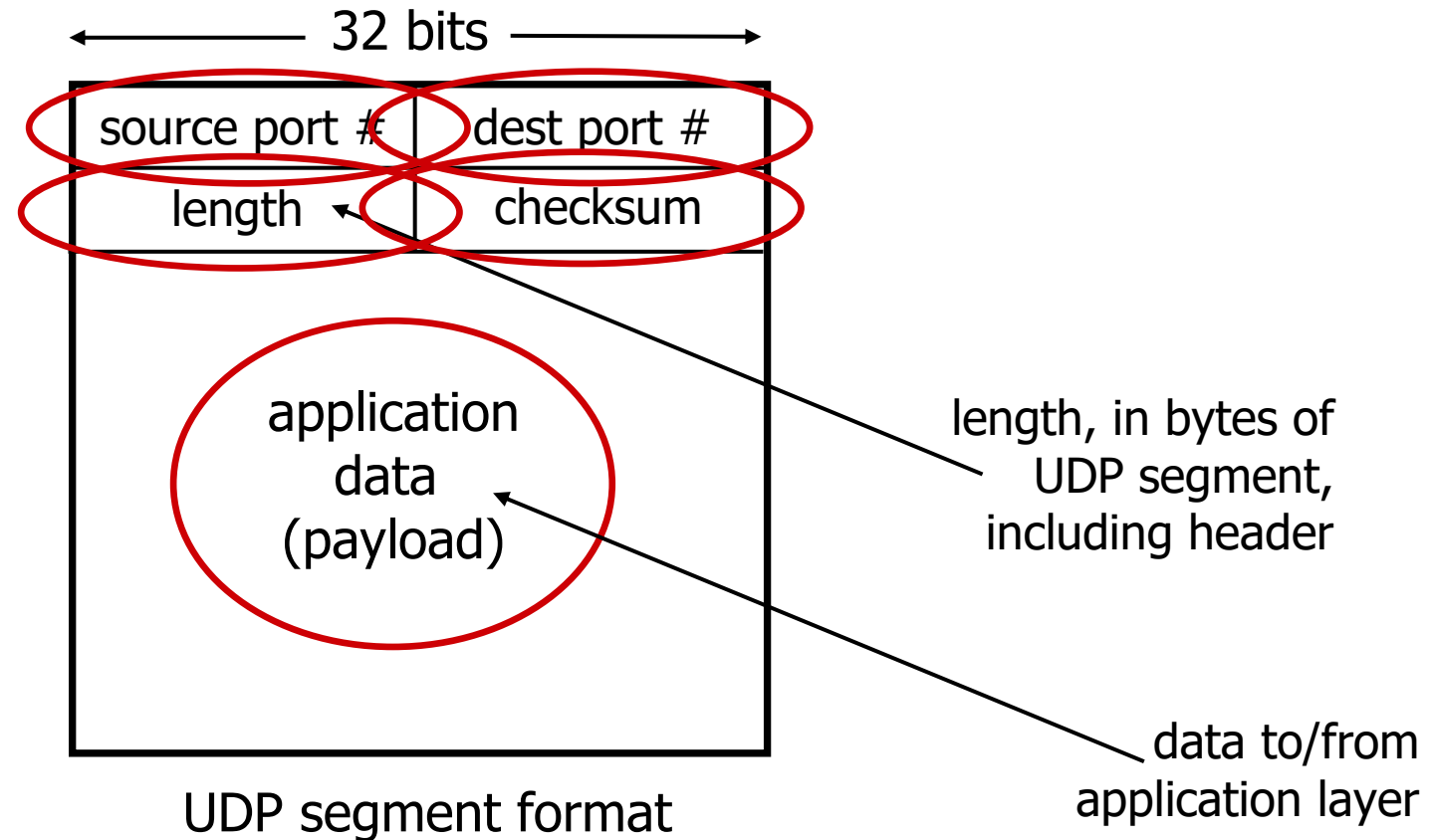
UDP: Transport Layer Actions



UDP: Transport Layer Actions

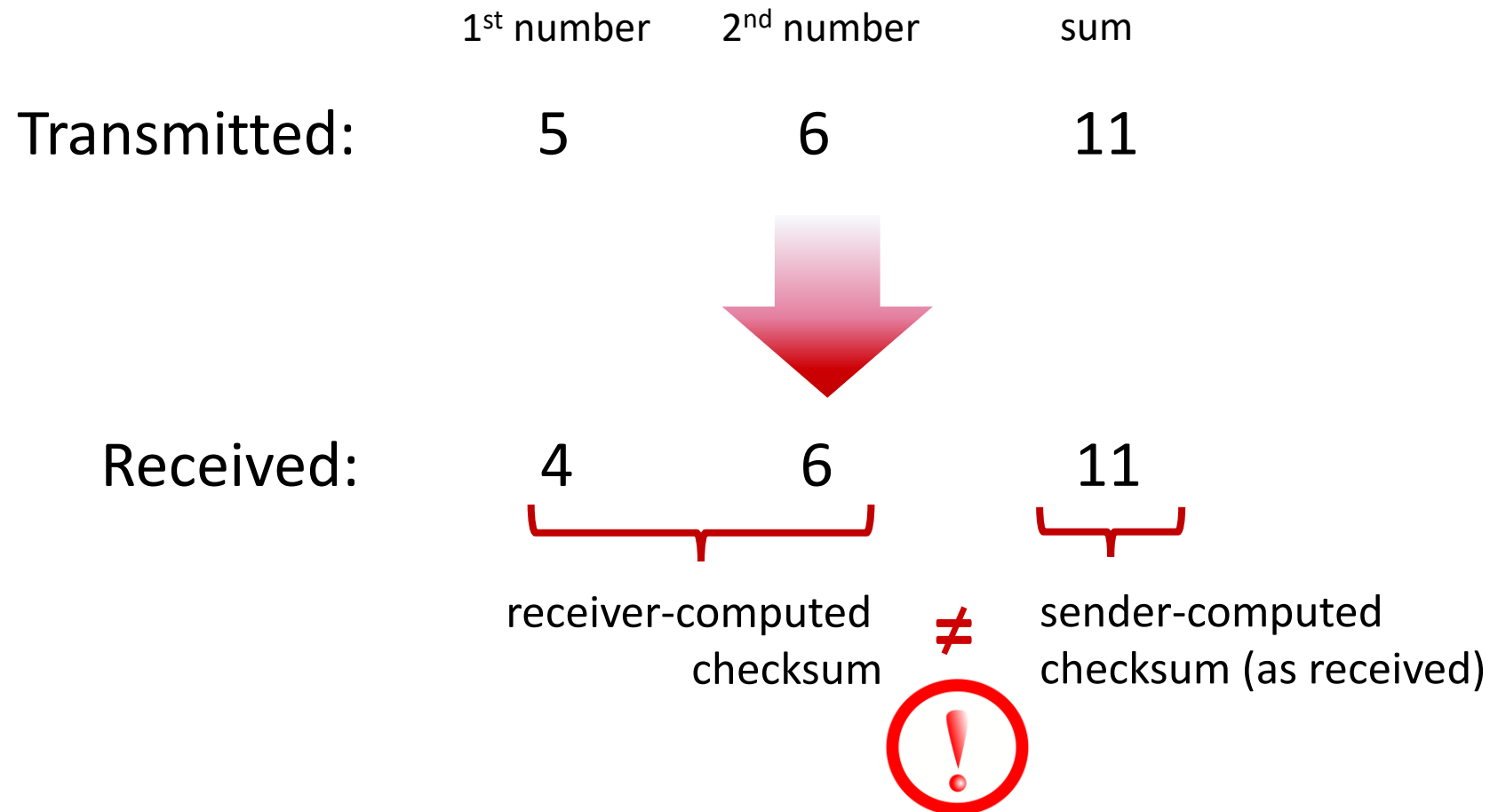


UDP segment header



UDP checksum

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



Internet 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

Internet checksum

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

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 |
| <hr/> | | | | | | | | | | | | | | | | |
| wraparound | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 1 |
| <hr/> | | | | | | | | | | | | | | | | |
| 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

| | | | | | | | | | | | | | | | | |
|------------|-------|---|---|---|---|---|---|---|---|---|---|---|---|---|---|---|
| | 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 |
| | <hr/> | | | | | | | | | | | | | | | |
| wraparound | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 1 | 1 | 1 | 0 | 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 |

Even though numbers have changed (bit flips), *no* change in checksum!

Summary: UDP

- “no frills” protocol:
 - segments may be lost, delivered out of order
 - best effort service: “send and hope for the best”
- UDP has its plusses:
 - no setup/handshaking needed (no RTT incurred)
 - can function when network service is compromised
 - helps with reliability (checksum)
- build additional functionality on top of UDP in application layer (e.g., HTTP/3)

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Interactive Problem

- Compute the Internet checksum value for these two 16-bit words:

11100010 11000111
*this binary number is 58055
decimal (base 10)*

11011110 11111010
*this binary number is 57082
decimal (base 10)*

1. What is the sum of these two 16 bit numbers?
2. Using the sum from question 1, what is the checksum?

Answer

1.

11100010 11000111

11011110 11111010

1 11000001 11000001

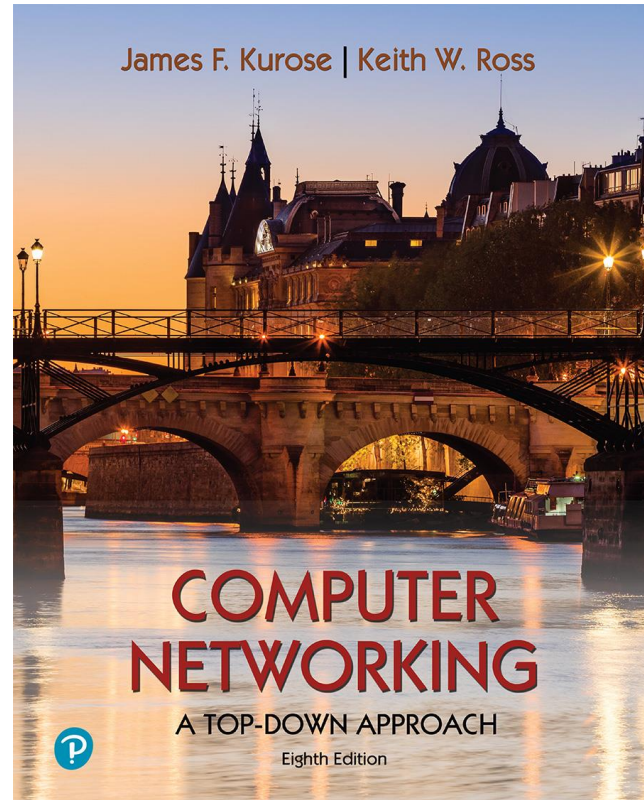
1

11000001 11000010

2.

00111110 00111101

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8th edition

Jim Kurose, Keith Ross

Pearson, 2020

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