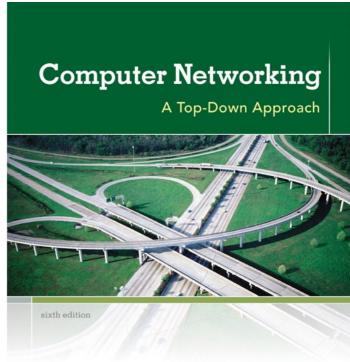
CN-Basic L19

User Datagram Protocol

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Chapter 3 Transport Layer



KUROSE ROSS

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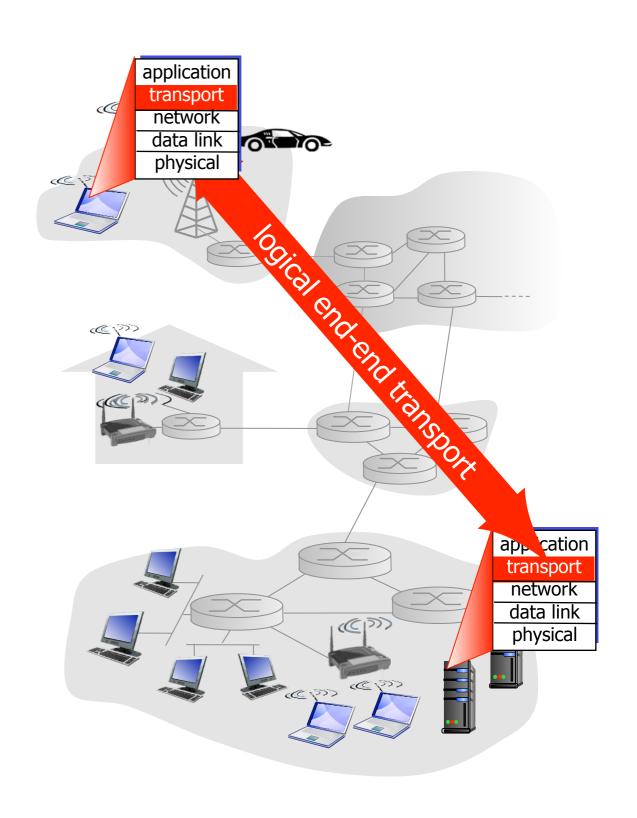
Chapter 3: Transport Layer

- Goals:
- 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: connectionoriented reliable transport

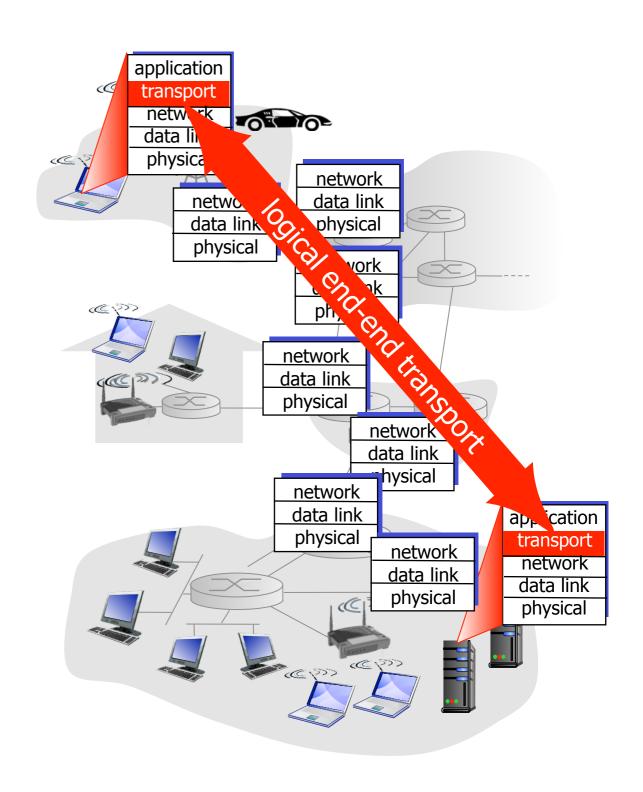
Transport services and protocols

- Provides logical communication between app processes running on different hosts
- Transport protocols run in end systems
 - Send side: breaks app messages into segments, passes to network layer
 - Rcv side: reassembles segments into messages, passes to app layer
- More than one transport protocol available to apps
 - Internet: TCP and UDP



Internet transport-layer protocols

- Reliable, in-order delivery (TCP)
 - Congestion control
 - Flow control
 - Connection setup
- Unreliable, unordered delivery: UDP
 - No-frills extension of "best-effort" IP
- Services not available:
 - Delay guarantees
 - Bandwidth guarantees



Transport layer protocol

- How would you design it
- What would you like to achieve
 - At simplest level
 - •Multiplex/de-multiplex
 - At advanced level
 - Reliable delivery i.e. Data integrity
 - Include error detection and retransmissions
 - Sequential delivery
 - Would need buffer
 - Detecting packet loss or duplicate delivery
 - Message boundaries
 - Security

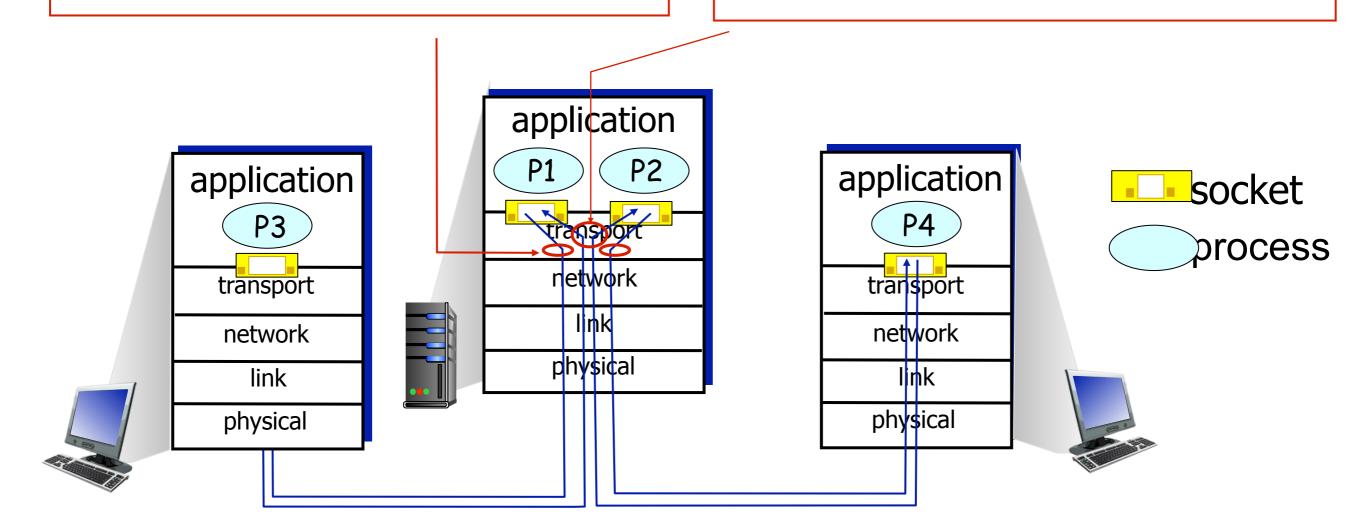
Multiplexing/demultiplexing

multiplexing at sender:

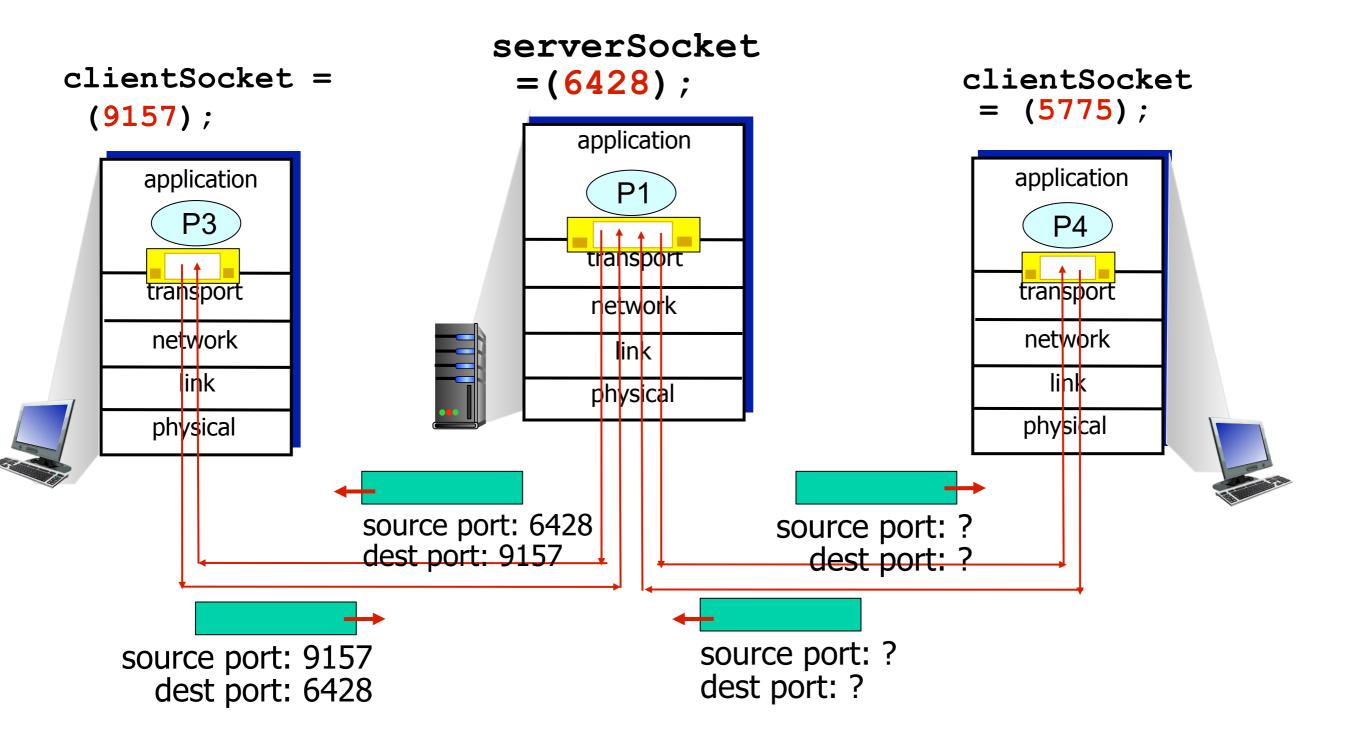
handle data from multiple sockets, add transport header (later used for demultiplexing)

demultiplexing at receiver:

use header info to deliver received segments to correct socket



Connectionless demux: example

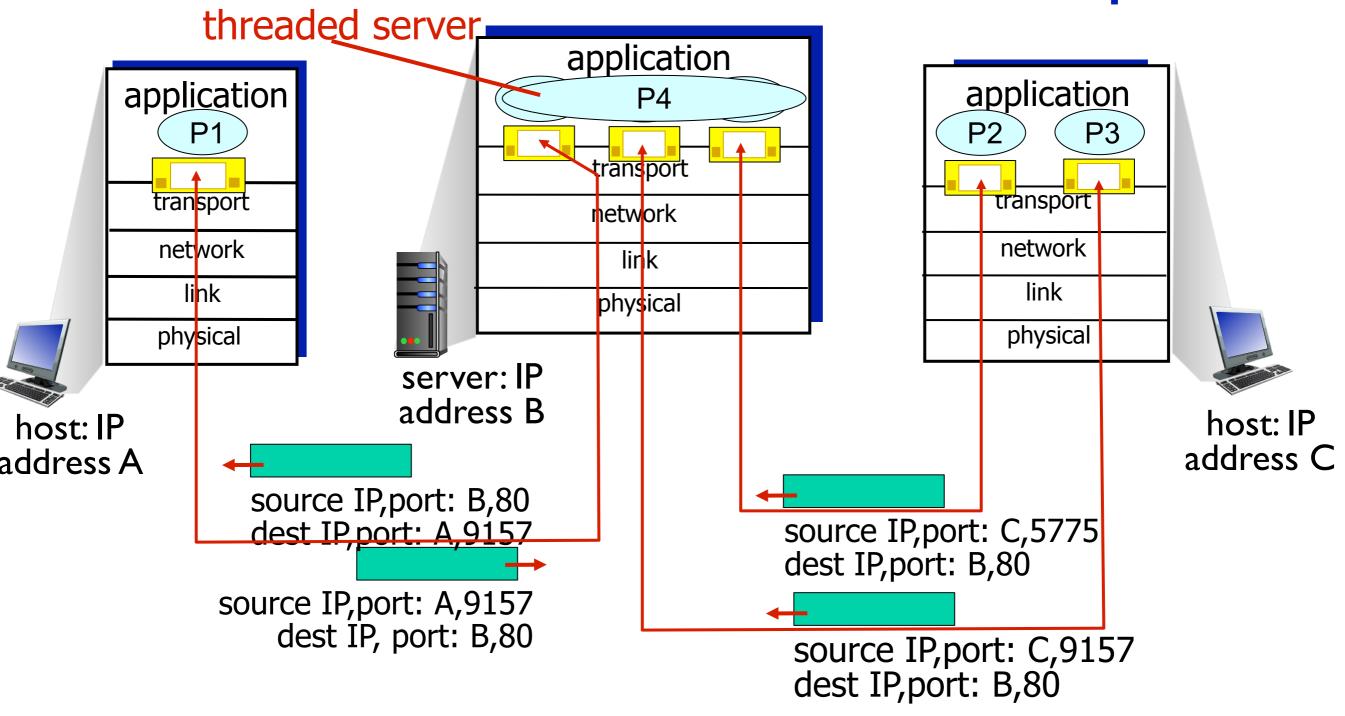


Connection-oriented demux

- Transport layer socket identified by 4(or 5)tuple:
 - source IP address
 - source port number
 - dest IP address
 - dest port number
 - (Protocol (TCP))
- demux: receiver 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: example



three segments, all destined to IP address: B, dest port: 80 are demultiplexed to different sockets

Connectionless vs Connection oriented

- Connection less
 - Packets are not numbered
 - Packets may arrive out of order
 - No acknowledgement
 - Packets may be lost
 - No prior handshake
- Connection oriented
 - Setup, data xfer, and teardown phase
 - Provides Reliability
 - Ordered Delivery
 - Handles Error Control better

Reliability

- Reliable protocol
 - Needs error and flow control
 - Compels slower service
 - Require extra overheads
- Unreliable protocol
 - No extra overheads
- Reliability at data link layer
 - Provides error and flow control
 - Why do we need it at Transport layer?

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

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

UDP

- When to prefer UDP over TCP
 - Real time apps do not want congestion control
 - Some packet loss is okay
 - Connection handshake not required
 - No overhead and quick response e.g. DNS
 - analogy: SMS vs phone call, Alerts?
 - No connection state maintenance
 - OS has less resources overhead for TCP
 - Can support more UDP clients than TCP
 - Better utilisation efficiency
 - UPD overhead is 8 bytes vs 20 (min.) bytes of TCP

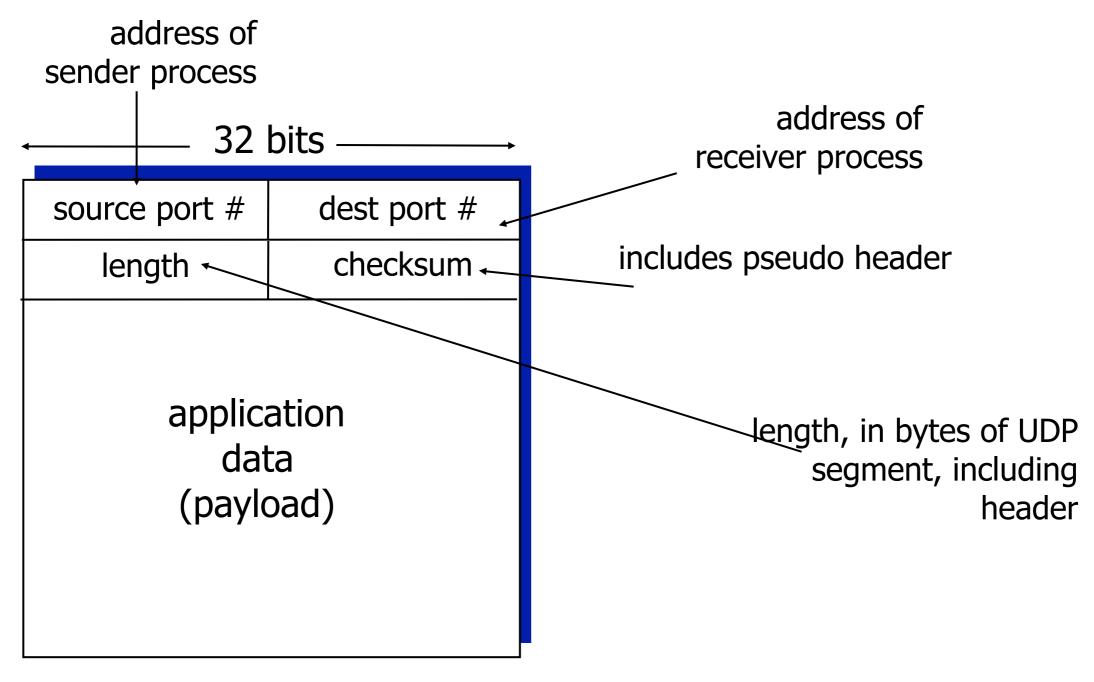
UDP

- How will you design a simple transport layer ?
- Just provides transport on top of IP
 - Multiplexing and demultiplexing
 - Little bit of error checking
 - No handshake
- Rest all has to be managed by application
 - Application practically talks to IP
- DNS uses UDP
 - What happens when query/response is lost?

UDP Headers

- What it should contain
 - Destination port number
 - Delivery to destination application process
 - Source port number. Why?
 - In case response needs to be returned back
 - Receiver identifies the sender's receiving point
 - Length
 - Each message can be of different length
 - Checksum
 - To detect if packet is corrupted

UDP: segment header



UDP segment format

Internet checksum: 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: example

Consider 3 words

```
■0110 0110 0110 0000 - 0x6660
```

```
■0101 0101 0101 0101 - 0x5555
```

```
■1000 1111 0000 1100 - 0x8F0C
```

- 10100 1010 1100 0001 -0x14AC1
- Wrapping around the overflow bit makes it
 - ■0100 1010 1100 0010 0x4AC2
- 1's complement will be
 - ■1011 0101 0011 1101 0xB53D

UDP checksum

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

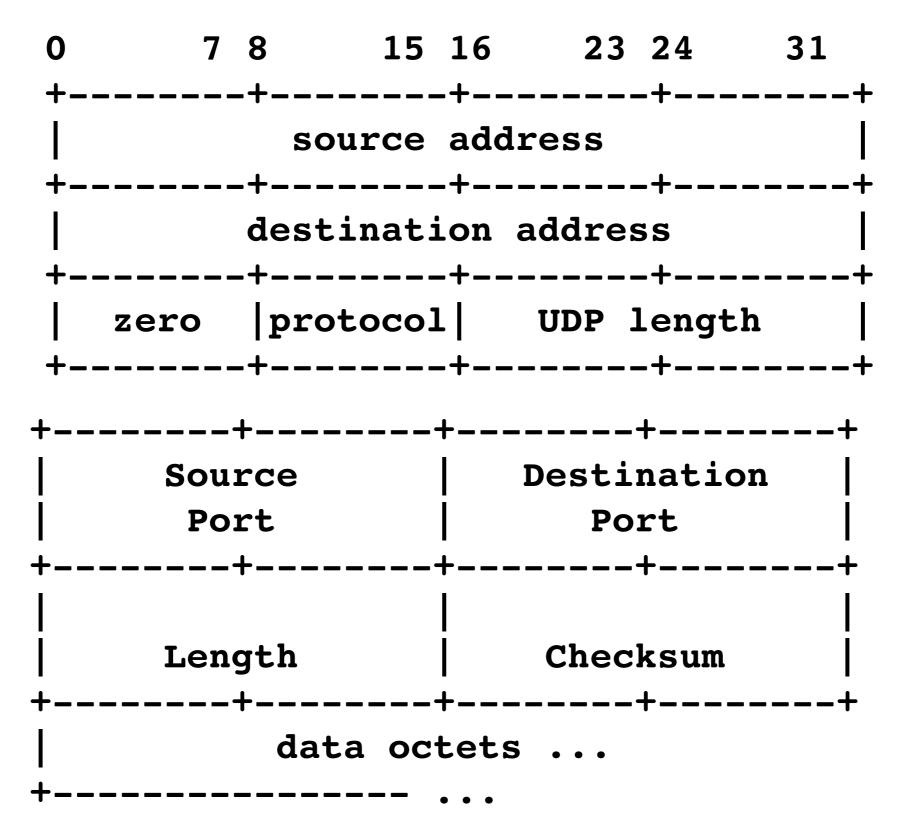
sender:

- treat segment contents, including header fields, as sequence of 16-bit integers
- checksum: addition (one'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

Pseudo header for checksum



UDP headers

- Why pseudo headers ?
 - Protection against misrouted datagrams
- When computed checksum is zero
 - Transmitted as all ones
 - (equivalent to 0 in 1's complement)
 - All zero checksum implies no checksum generated
- Checksum
 - Uses pseudo header, UDP header and data
- Length: min value is 8 (why?)
- Data: padded if needed
 - to make a multiple of 16 bits octets

UDP Checksum Exercise

- Compute Checksum for the following case
 - Src IP: 10.30.26.1, Dest IP: 10.30.26.11
 - **Src Port**: 16384, **Dest port**: 53
 - Application Data: "TESTING"
 - ■0x 54 45 53 54 49 4E 47
 - Hint: Do you need padding?
 - Define Pseudo-header, UDP Protocol value is 0x11
- Answer
 - •0A14 + 1001 + 0A14 + 110B + 0011 + 000F +
 - **4**000 + 0035 + 000F + 5445 + 5354 + 494E +
 - **4**700 **=** ??
- What is checksum when data is
 - "UQUQUQTESTING" or
 - "INSTTEG"

Summary

- Transport Protocol
- Multiplexing and Demultiplexing
- Connectionless and Connection oriented
- UDP protocol
- UDP Checksum