### Network Programming

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## Transport Layer UDP Protocol

## Transport Layer

#### Our goals:

- understand principles behind transport layer services:
  - multiplexing/demultipl exing
  - reliable data transfer
  - o flow control
  - congestion control

- learn about transport layer protocols in the Internet:
  - UDP: connectionless transport
  - TCP: connection-oriented transport
  - TCP congestion control

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## Transport services and protocols

- provide *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



### Transport vs. network layer

*network layer:* logical communication between hosts
 *transport layer:* logical communication between processes

• relies on, enhances, network layer services



# Transport Protocols





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# 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:
  - o delay guarantees
  - o bandwidth guarantees



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#### IP addresses versus port numbers



#### IANA ranges



#### Port Numbers in UNIX

In UNIX, the well-known ports are stored in a file called /etc/services. Each line in this file gives the name of the server and the well-known port number. We can use the grep utility to extract the line corresponding to the desired application. The following shows the port for FTP. Note that FTP can use port 21 with either UDP or TCP.

\$ grep	ftp	/etc/services
ftp	21/tcp	
ftp	21/udp	

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#### Socket address



### Multiplexing/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



## Connectionless demultiplexing

#### When host receives UDP Create sockets with port segment: numbers: checks destination port DatagramSocket mySocket1 = new number in segment DatagramSocket( ); o directs UDP segment to DatagramSocket mySocket2 = new socket with that port DatagramSocket(12535); number UDP socket identified by IP datagrams with two-tuple: different source IP (dest IP address, dest port number) addresses and/or source port numbers directed to same socket

# Connectionless demux (cont)

P2 Ρ3 SP: 6428 SP: 6428 DP: 5775 DP: 9157 SP: 9157 SP: 5775 DP: 6428 DP: 6428 client Client server IP:B TP: A IP: C SP provides "return address"

DatagramSocket serverSocket = new DatagramSocket(6428);

### Connection-oriented demux

- TCP socket identified by 4-tuple:
  - source IP address
  - source port number
  - o 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

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# <u>Connection-oriented demux</u> (cont)



## <u>Connection-oriented demux:</u> <u>Threaded Web Server</u>



#### Position of UDP, TCP, and SCTP in TCP/IP suite



# OSI Model



Figure 1.14 Layers on OSI model and Internet protocol suite

- Why do both sockets and XTI provide the interface from the upper three layers of the OSI model into the transport layer?
  - First, the upper three layers handle all the details of the application and The lower four layers handle all the communication details.
  - Second, the upper three layers is called a user process while the lower four layers are provided as part of the operating system kernel.

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#### 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

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#### 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
- often used for streaming multimedia apps
  >loss tolerant
  - ➤rate sensitive
- · other UDP uses  $\rightarrow$  DNS and SNMP
- Suitable for multicasting

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### UDP: more



## UDP checksum

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

#### <u>Sender:</u>

- 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



# UDP output





#### Queues in UDP



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# UDP is suitable for:

- A process that requires simple requestresponse communication with little concern for flow and error control - FTP
- A process with internal flow and error control mechanisms - TFTF
- Multicasting
- Management processes SNMP
- □ Some rout updating protocols RIP