CS2105

An **Awesome** Introduction to Computer Networks

Lectures 2&3: The Application Layer



Packet Switching



- The Internet is a packet switching network
 - Hosts share and contend network resources.
 - Application message is broken into a bunch of packets and sent onto the link one by one.
 - A router stores and forwards packets.

Receiver assembles all the packets to restore the

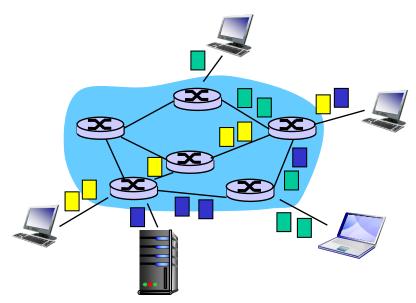
application message.

Anyone can use the network at any time

Bandwidth division into pieces"

Dedicated allocation

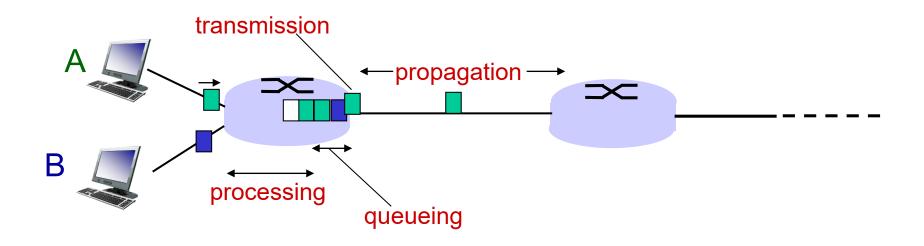
Resource reservation



PREVIOUS LECTURE

Packet Delay

- End-to-end delay is the time taken for a packet to travel from source to destination. It consists of:
 - processing delay
 - queueing delay
 - ! Transmission delay
 - Propagation delay

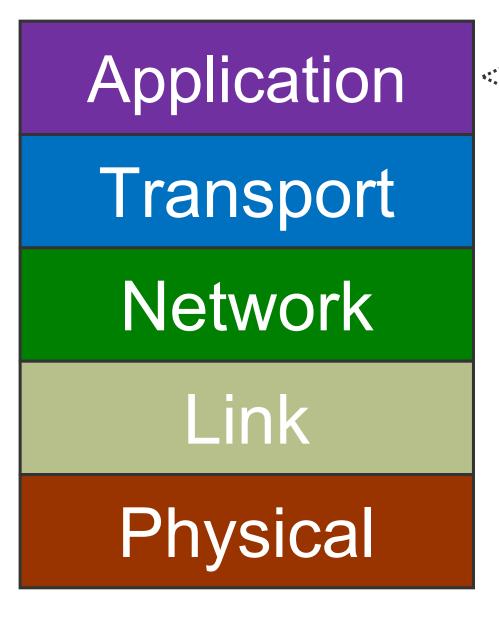


PREVIOUS LECTURE

Network Protocols

Networks are complex. There are many issues to consider, to support different applications running on large number of hosts through different access technology and physical media.

- Protocols regulate communication activities in a network.
 - Define the *format* and *order* of messages exchanged between hosts for a specific purpose.



You are here

Upper layer protocols will use the service provided by lower layers

Lectures 2&3: Application Layer

After this class, you are expected to:

- understand the basic HTTP interactions between the client and the server, the concepts of persistent and non-persistent connections.
- understand the services provided by DNS and how a DNS query is resolved.
- understand the concept of socket.
- be able to write simple client/server programs through socket programming.

Lectures 2&3: Roadmap

- 2.1 Principles of Network Applications
- 2.2 Web and HTTP
- **2.4 DNS**
- 2.7 Socket programming



Kurose Textbook, Chapter 2 (Some slides are taken from the book)

Evolution of Network Applications

- Early days of Internet
 - Remote access (e.g. telnet, now ssh)
- ❖ 70s 80s
 - Email, FTP
- ◆ 90s
 - Web
- ❖ 2000s now
 - P2P file sharing
 - Online games
 - Instant Messaging, Skype
 - YouTube, Facebook

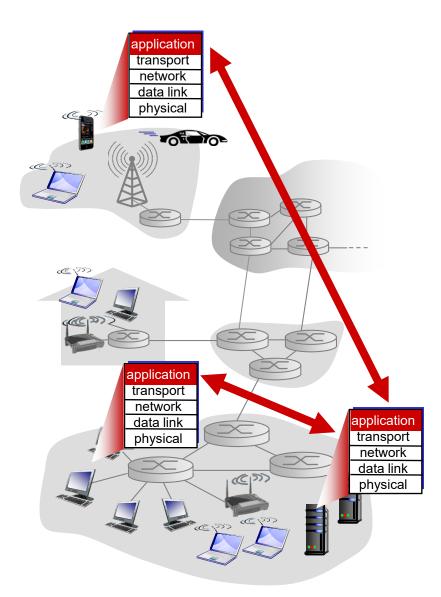
Creating Network Applications

write programs that:

- run on (different) hosts
- communicate over network
- ◆ e.g., web server software⇒ browser software

classic structure of network applications:

- Client-server
- Peer-to-peer (P2P)



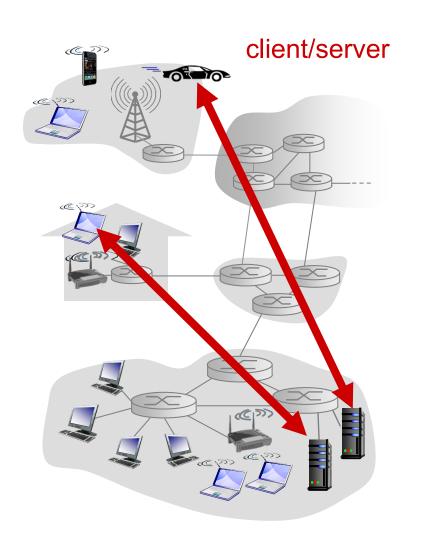
Client-Server Architecture

Server:

- Waits for incoming requests
- Provides requested service to client
- data centers for scaling

Client:

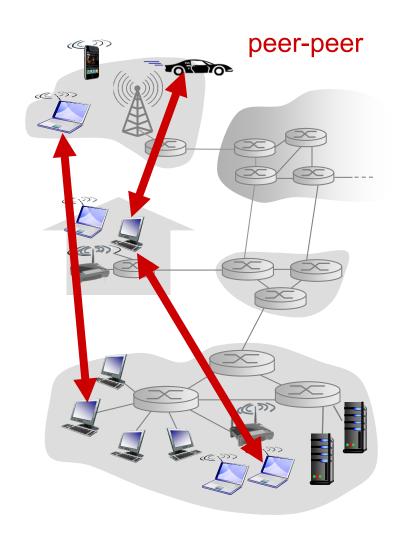
- Initiates contact with server ("speaks first")
- Typically requests service from server
- For Web, client is usually implemented in browser



P2P Architecture

download and upload simultaneously

- No always-on server
- Arbitrary end systems directly communicate.
- Peers request service from other peers, provide service in return to other peers
 - self scalability new peers bring new service capacity, as well as new service demands
- Peers are intermittently connected and change IP addresses
 - complex management



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What transport service does an app need?

retransmission is not useful since the old/corrupted data might not be useful anymore

Data integrity

- some apps (e.g., file transfer, web transactions) require
 100% reliable data transfer
- other apps (e.g., audio streaming) can tolerate some data loss

Timing

 some apps (e.g., online interactive games) require low delay to be "effective"

Throughput

- some apps (e.g., multimedia) require minimum amount of bandwidth to be "effective"
- other apps (e.g., file transfer) make use of whatever throughput available

Security

encryption, data integrity, authentication ...

Requirements of Example Apps

Application	Data loss	Throughput	Time-sensitive
File transfer	No loss	Elastic	No
Electronic mail	No loss	Elastic	No
Web documents	No loss	Elastic	No
Real-time audio/video	Loss-tolerant	Audio: 5kbps-1Mbps	Yes: 100s of msec
		Video:10kbps-5Mbps	
Stored audio/video	Loss-tolerant	Same as above	Yes: few seconds
Interactive games	Loss-tolerant	Few kbps – 10 kbps	Yes: 100s of msec
Text messaging	No loss	Elastic	Yes and no
		·	

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App-layer Protocols Define...

- types of messages exchanged
 - e.g., request, response
- message syntax:
 - what fields in messages & how fields are delineated
- message semantics
 - meaning of information in fields
- rules for when and how applications send & respond to messages

- open protocols:
 - defined in RFCs
 - allows for interoperability
 - e.g., HTTP, SMTP

- proprietary protocols:
 - e.g., Skype

Transport Layer Protocols: TCP/UDP

App-layer protocols ride on transport layer protocols:

TCP service:

- <u>reliable</u> data transfer
- * flow control: sender won't overwhelm receiver
- congestion control: throttle sender when network is overloaded
- does not provide: timing, minimum throughput guarantee, security

UDP service:

- unreliable data transfer
- no flow control
- no congestion control

 does not provide: timing, throughput guarantee or security

cannot guarantee any performance/throughput because the packet switching network does not allow for any reservation of the resource - inherent problem of internet architecture

Example App/Transport Protocols

Application	Application Layer Protocol	Underlying Transport Protocol
Electronic mail	SMTP [RFC 5321]	TCP
Remote terminal access	Telnet [RFC 854]	TCP
Web	HTTP [RFC 2616]	TCP
File transfer	FTP [RFC 959]	TCP
Streaming multimedia	HTTP (e.g., YouTube)	Typically TCP
	SIP [RFC 3261],	
Internet telephony	RTP [RFC 3550],	
	or proprietary	TCP or UDP
	(e.g., Skype)	

Lectures 2&3: Roadmap

- 2.1 Principles of Network Applications
- 2.2 Web and HTTP
- **2.4 DNS**
- 2.7 Socket programming

The Web: Some Jargon

- A Web page typically consists of:
 - base HTML file, and
 - several referenced objects.
- An object can be HTML file, JPEG image, Java applet, audio file, ...
- Each object is addressable by a URL, e.g.,

www.comp.nus.edu.sg/~cs2105/img/doge.jpg
host name path name

HTTP Overview

HTTP: <u>Hypertext transfer</u> <u>protocol</u>

- Web's application layer protocol
- Client/server model
 - client: usually is browser that requests, receives and displays Web objects
 - server: Web server sends objects in response to requests
- http 1.0: RFC 1945
- http 1.1: RFC 2616



HTTP uses TCP as transport service

- Client initiates TCP connection to server.
- Server accepts TCP connection request from client.
- HTTP messages are exchanged between browser (HTTP client) and Web server (HTTP server) over TCP connection.
- TCP connection closed.

Two Versions of HTTP

Non-persistent HTTP

- at most one object sent over a TCP connection
 - connection then closed
- downloading multiple objects required multiple connections

Persistent HTTP

multiple objects can
be sent over single
TCP connection
between client, server

Non-persistent HTTP Example

suppose user enters URL:

www.comp.nus.edu.sg/~cs2105/demo.html

(contains text, reference to a jpeg image)

- 1a. HTTP client initiates TCP connection to HTTP server at www.comp.nus.edu.sg on port 80
- 1b. HTTP server at host www.comp.nus.edu.sg is waiting for TCP connection at port 80.It "accepts" connection and reply client

- HTTP client sends HTTP
 request message (containing URL) into TCP connection socket. Message indicates
 that client wants object
 ~cs2105/demo.html
- 3. HTTP server receives request message, forms response message containing requested object, and sends message to the client

time

time

Non-persistent HTTP Example



displays html. Parsing html
tile, client notices the

referenced jpeg object

Steps 1-5 repeated for the jpeg object 4. HTTP server closes TCP connection.

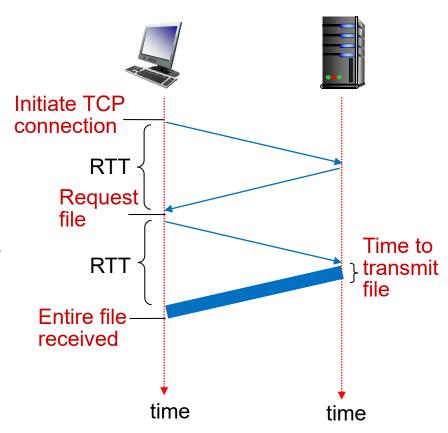
- This is an example of non-persistent connection (http 1.0).
 - One object per connection
- HTTP 1.1 allows persistent connection (to discuss).
 - Multiple objects per connection

Non-persistent HTTP: Response Time

time for a packet to travel from client to server and go back

HTTP response time:

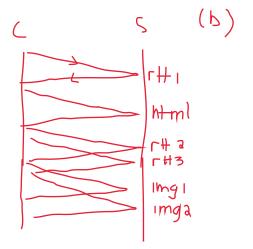
- one RTT to establish TCP connection
- one RTT for HTTP request and the first few bytes of HTTP response to return
- file transmission time
- non-persistent HTTP response time =
 - 2 * RTT+ file transmission time



Persistent HTTP

non-persistent HTTP issues:

- requires 2 RTTs per object
- OS overhead for each TCP connection
- browsers often open parallel
 TCP connections to fetch
 referenced objects



persistent HTTP:

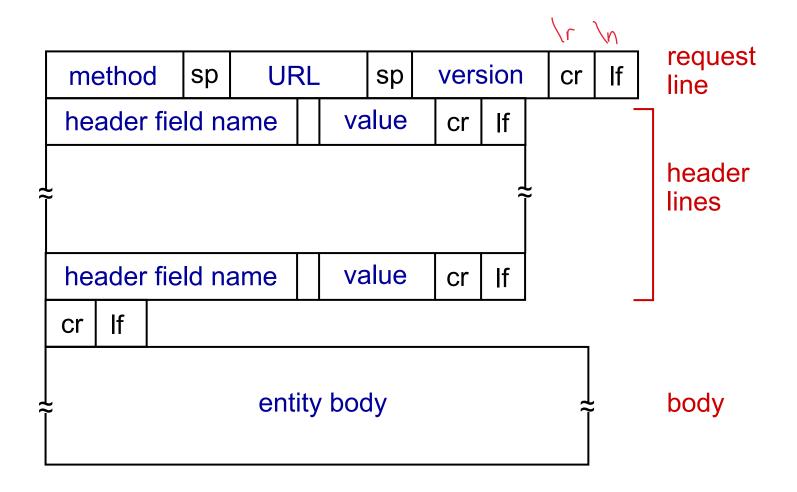
- server leaves connection open after sending response
- subsequent HTTP messages between same client/server sent over the same TCP connection
- moreover, client may send requests as soon as it encounters a referenced object (persistent with pipelining)
 - as little as one RTT for all the referenced objects

Example HTTP Request Message

- Two types of HTTP messages: request, response
- HTTP request message:

```
request line
(GET method)
                /index.html HTTP/1.1\r\n
            Host: www.example.org\r\n
    header
            Connection: keep-alive\r\n
     lines
                     Extra "blank" line indicates
                     the end of header lines
```

HTTP Request Message: General Format



HTTP Request Method Types

HTTP/1.0:

- GET
- POST
 - web page often includes form input
 - input is uploaded to server in entity body

HEAD

 asks server to leave requested object out of response

HTTP/1.1:

- GET, POST, HEAD
- PUT
 - uploads file in entity body to path specified in URL field
- DELETE
 - deletes file specified in the URI field

Example HTTP Response Message

```
status line
(protocol status code)
                       client request is successful - shows the response to the request here
         HTTP/1.1 200 OK\r\n
         Date: Wed, 23 Jan 2019 13:11:15 GMT\r\n
         Content-Length: 606\r\n
header
         Content-Type: text/html\r\n
  lines
                                                        data, e.g.
          r\n
                                                        requested
                                                        HTML file
         data data data data
```

For a full list of request/response header fields, check http://www.w3.org/Protocols/rfc2616/rfc2616-sec14.html

HTTP Response Status Codes

- ❖ Status code appears in 1st line in server-to-client response message.
- Some sample codes:

200 OK

request succeeded, requested object later in this msg

301 Moved Permanently

 requested object moved, new location specified later in this msg (Location:)

403 Forbidden

server declines to show the requested webpage

404 Not Found

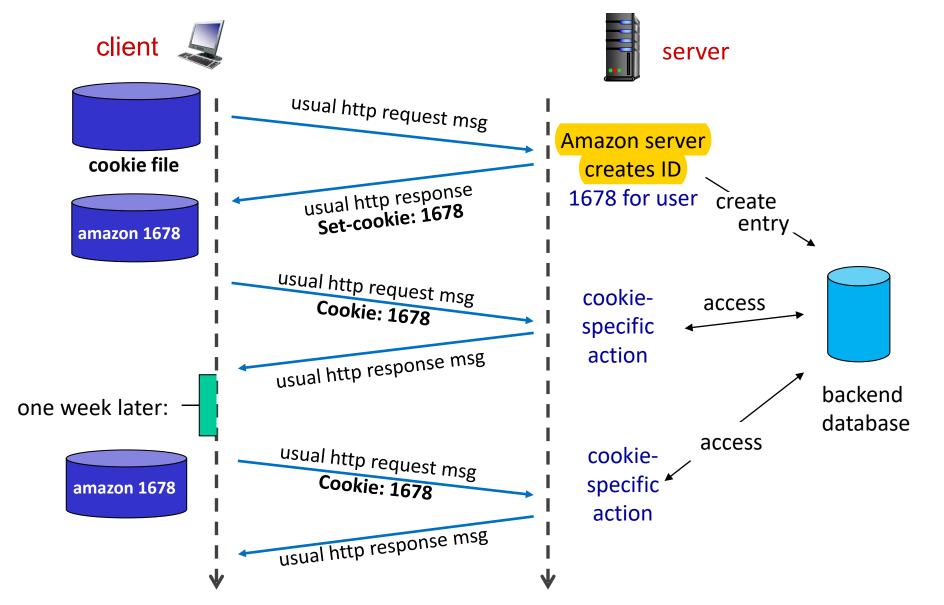
requested document not found on this server

For a full list of status codes, check http://www.w3.org/Protocols/rfc2616/rfc2616-sec10.html

Cookies

- HTTP is designed to be "stateless".
 - Server maintains no information about past client requests.
- Sometimes it's good to maintain states (history) at server/client over multiple transactions.
 - E.g. shopping carts
- Cookie: http messages carry "state"
 - 1) cookie header field of HTTP request / response messages can be used as a token to prove identity
 - cookie file kept on user's host, managed by user's browser
 - 3) back-end database at Web site

Keep User States with Cookie



Conditional GET

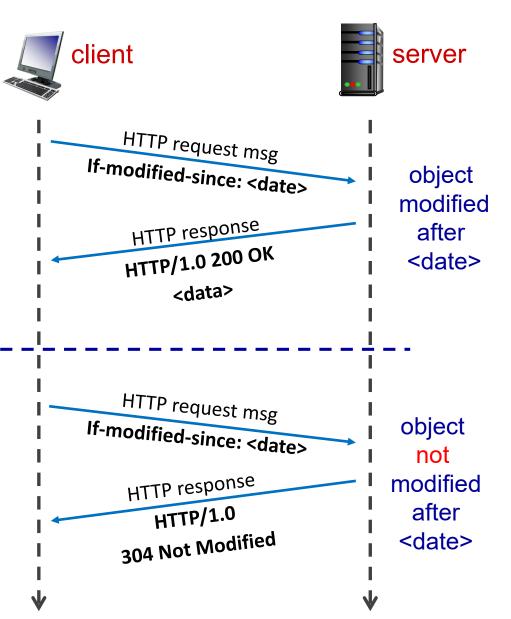
- Goal: don't send object if (client) cache has up-todate cached version
- cache: specify date of cached copy in HTTP request

If-modified-since:
 <date>

server: response contains no object if cached copy is up-to-date:

HTTP/1.0 304 Not Modified

saves bandwidth



Lectures 2&3: Roadmap

- 2.1 Principles of Network Applications
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- **2.4 DNS**
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Domain Name System [RFC 1034, 1035]

- Two ways to identify a host:
 - **Hostname**, e.g., www.example.org
 - IP address, e.g., 93.184.216.34
- DNS (Domain Name System) translates between the two.
 - A client must carry out a DNS query to determine the IP address corresponding to the server name (e.g., www.example.org) prior to the connection.

```
cononical name = actual / true name
```

DNS: Resource Records (RR)

Mapping between host names and IP addresses (and others) are stored as <u>resource records</u> (RR).

RR format: (name, value, type, ttl)

<u>type = A</u>

- name is hostname
- value is IP address

type = NS name server

- name is domain (e.g., nus.edu.sg)
- value is hostname of authoritative name server for this domain

<u>type = CNAME</u>

- name is alias name (e.g. www.nus.edu.sg) for some "canonical" (the real) name
- value is canonical name (e.g. mgnzsqc.x.incapdns.net)

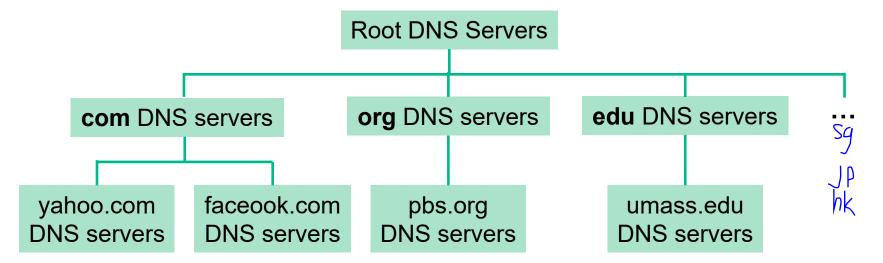
mail exchanger = email server

type = MX

value is name of mail server associated with name

Distributed, Hierarchical Database

DNS stored RR in distributed databases implemented in hierarchy of many name servers.

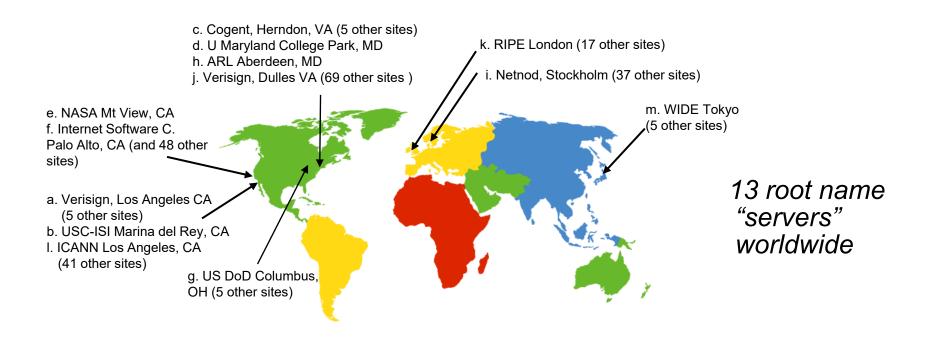


A client wants IP address for www.facebook.com:

- client queries root server to find .com DNS server
- client queries .com DNS server to get facebook.com DNS server
- client queries facebook.com DNS server to get IP address for www.facebook.com

Root Servers

Answers requests for records in the root zone by returning a list of the authoritative name servers for the appropriate top-level domain (TLD).



TLD and Authoritative Servers

Top-level domain (TLD) servers:

responsible for com, org, net, edu, ... and all toplevel country domains, e.g., uk, sg, jp

Authoritative servers:

- Organization's own DNS server(s), providing authoritative hostname to IP mappings for organization's named hosts (e.g. Web, mail)
- can be maintained by organization or service provider

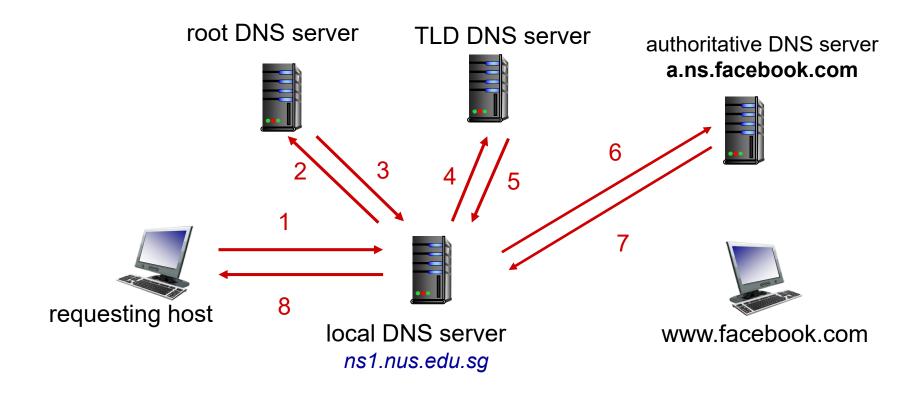
Local DNS Server

- Does not strictly belong to hierarchy
- Each ISP (residential ISP, company, university) has one local DNS server.
 - also called "default name server"
- When host makes a DNS query, query is sent to its local DNS server
 - Retrieve name-to-address translation from local cache
 - Local DNS server acts as proxy and forwards query into hierarchy if answer is not found locally

DNS Caching

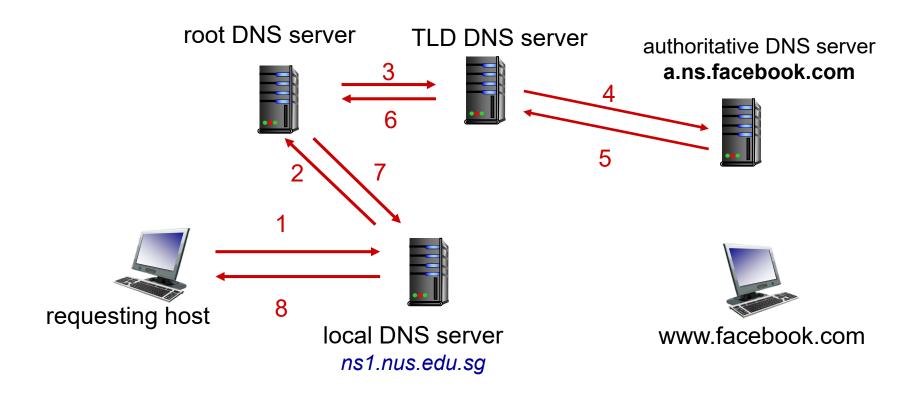
- ❖ Once a name server learns mapping, it *caches* mapping.
 - cached entries may be out-of-date (best effort name-toaddress translation!)
 - cached entries expire after some time (TTL).
 - if name host changes IP address, may not be known Internet-wide until all TTLs expire.
- Update/notify mechanisms proposed IETF standard
 - RFC 2136
- DNS runs over JDP.

DNS Name Resolution



This is known as iterative query.

DNS Name Resolution



- This is known as recursive query.
 - rarely used in practice

Lectures 2&3: Roadmap

- 2.1 Principles of Network Applications
- 2.2 Web and HTTP
- **2.4 DNS**
- 2.7 Socket programming: Creating Network Applications

Processes

- Process: program running within a host.
 - Within the same host, two processes communicate using inter-process communication (defined by OS).
 - Processes in different hosts communicate by exchanging messages (according to protocols).

In C/S model

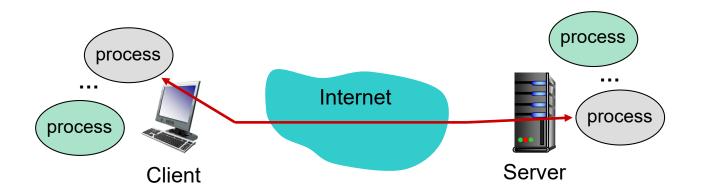
server process waits to be contacted

client process initiates the communication

Addressing Processes

- IP address is used to identify a host
 - A 32-bit integer (e.g. 137.132.21.27)
- Question: is IP address of a host suffice to identify a process running inside that host?

A: no, many processes may run concurrently in a host.



Addressing Processes

- A process is identified by (IP address, port number).
 - Port number is 16-bit integer (1-1023 are reserved for standard use).
- Example port numbers
 - HTTP server: 80
 - SMTP server: 25
- IANA coordinates the assignment of port number:
 - http://www.iana.org/assignments/service-names-portnumbers/service-names-port-numbers.xhtml

Analogy

Postal service:

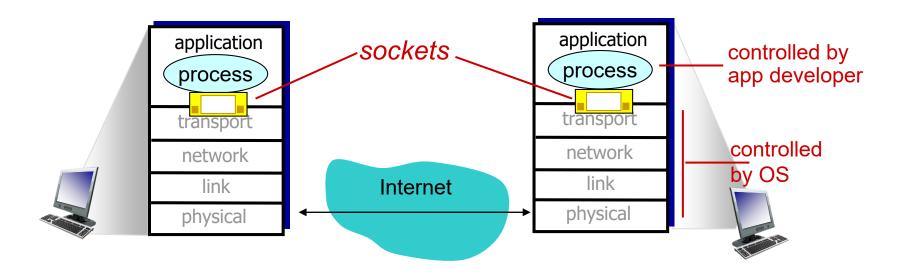
- deliver letter to the doorstep: home address
- dispatch letter to the right person in the house: name of the receiver as stated on the letter

Protocol service:

- deliver packet to the right host: IP address of the host
- dispatch packet to the right process in the host: port number of the process

Sockets

- Socket is the software interface between app processes and transport layer protocols.
 - Process sends/receives messages to/from its socket.
 - Programming-wise: a set of APIs



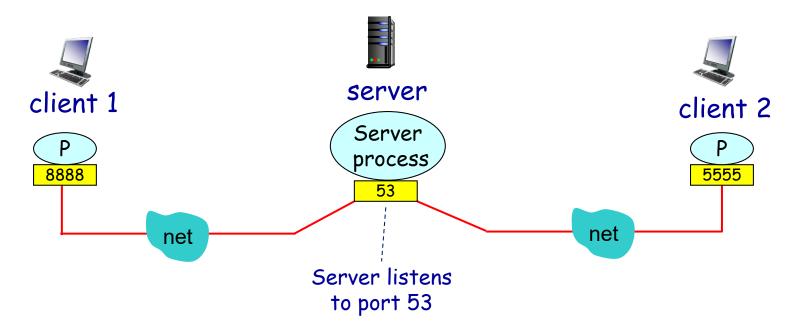
Socket Programming

- Applications (or processes) treat the Internet as a black box, sending and receiving messages through sockets.
- Two types of sockets
 - TCP: reliable, byte stream-oriented socket
 - UDP: unreliable datagram socket
- Now let's write a simple client/server application that client sends a line of text to server, and server echoes it.
 - We will demo both TCP socket version and UDP socket version.

Socket Programming with *UDP*

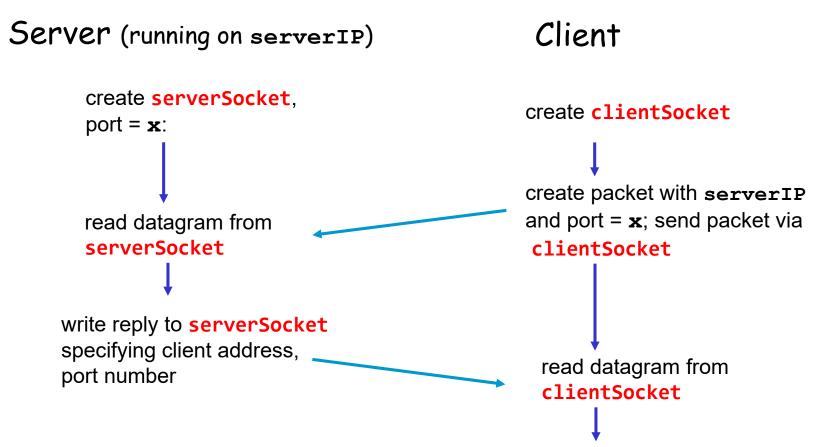
UDP: no "connection" between client and server

- Sender (client) explicitly attaches destination IP address and port number to <u>each packet</u>.
- Receiver (server) extracts sender IP address and port number from the received packet.



UDP: Client/server Socket Interaction

work flow



close clientSocket

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Example: UDP Echo Server

code for udp echo

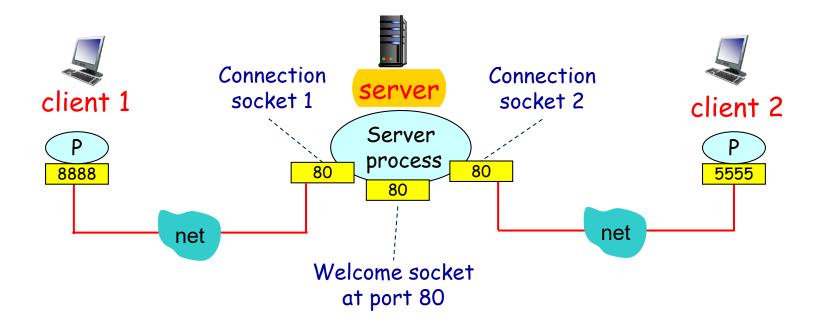
```
from socket import * — include Python's socket library
serverPort = 2105
                                       UDP socket
                            IPv4
# create a socket
serverSocket = socket(AF INET, SOCK DGRAM)
# bind socket to local port number 2105
serverSocket.bind(('', serverPort))
                                                 receive datagram
print('Server is ready to receive message')
                                                 buffer size: 2048B
# extract client address from received packet
message, clientAddress = serverSocket.recvfrom(2048)
serverSocket.sendto(message, clientAddress)
serverSocket.close()
```

Example: UDP Echo Client

```
from socket import *
serverName = 'localhost' # client, server on the same host
serverPort = 2105
clientSocket = socket(AF_INET, SOCK_DGRAM)
                                              convert message
message = input('Enter a message: ')
                                               from string to
                                               bytes and send it
# send msg to server address
clientSocket.sendto(message.encode(), (serverName, serverPort))
receivedMsg, serverAddress = clientSocket.recvfrom(2048)
print('from server:', receivedMsg.decode())
                                              convert from bytes
                                              to string
clientSocket.close()
```

Socket Programming with TCP

- When client creates socket, client TCP establishes a connection to server TCP.
- When contacted by client, server TCP creates a new socket for server process to communicate with that client.
 - allows server to talk with multiple clients individually.



TCP: Client/server Socket Interaction

Client Server (running on serverIP) create serverSocket, port = x wait for incoming TCP create clientSocket, connection request connection setup connect to serverIP, port = xconnectionSocket send request using clientSocket read request from connectionSocket write reply to read reply from clientSocket connectionSocket close connectionSocket close clientSocket

Example: TCP Echo Server

```
from socket import *
                                               TCP socket
serverPort = 2105
serverSocket = socket(AF INET, SOCK STREAM)
serverSocket.bind(('', serverPort))
                                         listens for incoming TCP request
                                         (not available in UDP socket)
serverSocket.listen()
print('Server is ready to receive message')
connectionSocket, clientAddr = serverSocket.accept()
message = connectionSocket.recv(2048)
                                                         returns a <u>new</u> socket
connectionSocket.send(message)
                                                         to communicate with
                   no need to specify the receiver here, since the connectionSocket
                                                         client socket
connectionSocket is specific for this client close()
```

Example: TCP Echo Client

```
from socket import *
serverName = 'localhost'
serverPort = 2105
clientSocket = socket(AF INET, SOCK STREAM)
                                                       establish a
clientSocket.connect((serverName, serverPort)) 
                                                       connection
message = input('Enter a message: ')
                                              no need to attach
clientSocket.send(message.encode())
                                              server name, port
receivedMsg = clientSocket.recv(2048)
print('from server:', receivedMsg.decode())
clientSocket.close()
```

TCP Socket vs. UDP Socket

- In TCP, two processes communicate as if there is a pipe between them. The pipe remains in place until one of the two processes closes it.
 - When one of the processes wants to send more bytes to the other process, it simply writes data to that pipe.
 - The sending process doesn't need to attach a destination IP address and port number to the bytes in each sending attempt as the logical pipe has been established (which is also reliable).
- In UDP, programmers need to form UDP datagram packets explicitly and attach destination IP address / port number to every packet.

Lectures 2&3: Summary

- Application architectures
 - Client-server
 - P2P
- Application service requirements:
 - reliability, throughput, delay, security

- Specific protocols:
 - HTTP
 - DNS

- Internet transport service model
 - connection-oriented, reliable: TCP
 - Connection-less, unreliable: UDP

Lectures 2&3: Summary

Socket programming

TCP socket

- When contacted by client, server TCP creates new socket.
- Server uses (client IP + port #) to distinguish clients.
- When client creates its socket, client TCP establishes connection to server TCP.

UDP socket

- Server uses one socket to serve all clients.
- No connection is established before sending data.
- Sender explicitly attaches destination IP address and port # to each packet.
- Transmitted data may be lost or received out-of-order.