UNIVERSITY OF COPENHAGEN



Network Apps: Overview of Socket API

Network Apps: HTTP & Content Delivery

David Marchant

Based on slides compiled by Marcos Vaz Salles, with adaptiona by Vivek Shah and Michal Kirkedal Thomsen

Exam / reexam weeks changes

- Week 45
 - Monday lecture and exercises as per normal
 - No lecture, exercises or cafes Wednesday 8th
 - Lecture Friday 10th 10.15→12, Location TBD
 - Cafe on Friday as per normal
- Week 46
 - No lectures or exercises all week
 - Cafes at regular times, Location TBD

Will update as soon as we can get locations confirmed



Recap: Key Concepts in Networking

Protocols

- Speaking the same language
- Syntax and semantics

Layering

- Standing on the shoulders of giants
- A key to managing complexity

Resource allocation

- Dividing scarce resources among competing parties
- Memory, link bandwidth, wireless spectrum, paths

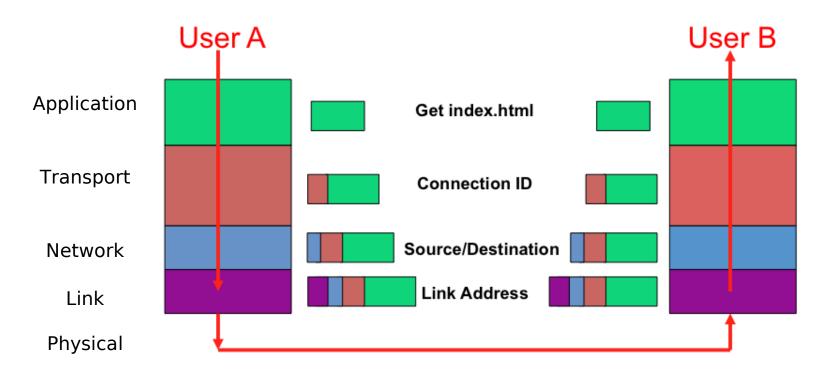
Naming

What to call computers, services, protocols, ...



Layer Encapsulation in HTTP

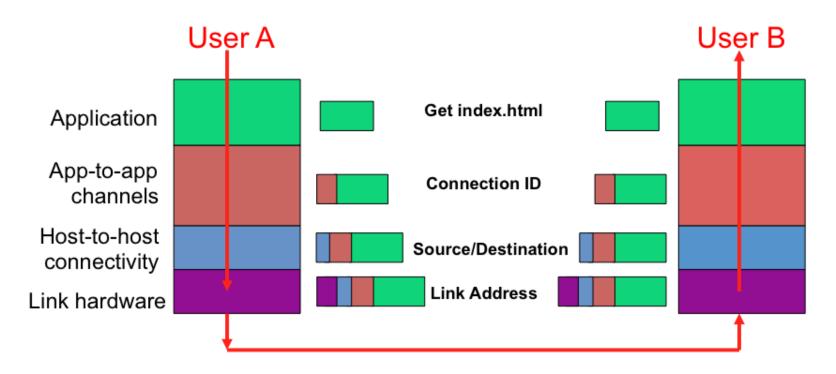






Layer Encapsulation in HTTP







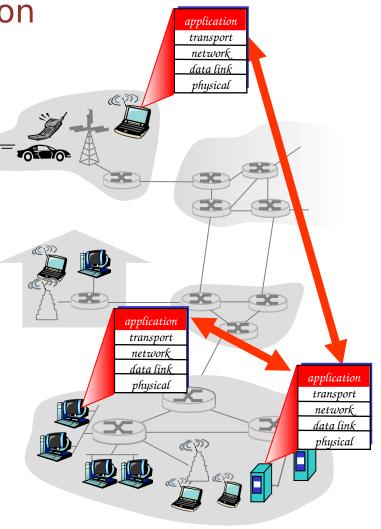
Creating a network application

write programs that

- run on (different) end systems
- communicate over network
- e.g., web server software communicates with browser software

No need to write software for network-core devices

- network-core devices do not run user applications
- applications on end systems allows for rapid app development, propagation





Source: Kurose & Ross

Some network apps

- e-mail
- remote login
- web
- instant messaging
- P2P file sharing
- multi-user network games
- streaming stored video (YouTube)

- voice over IP
- real-time video conferencing
- social networking
- cloud computing
- ...
- ...

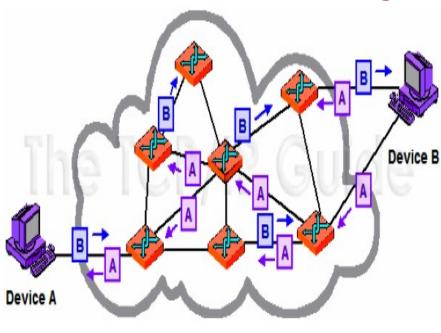


Network applications need streams of data

Circuit switching







Today's networks provide packet delivery, not streams!

Source: Freedman (partial)

What if the Data Doesn't Fit?

GET /courses/archive/spr09/cos461/ HTTP/1.1

Host: www.cs.princeton.edu

User-Agent: Mozilla/4.03

CRLF

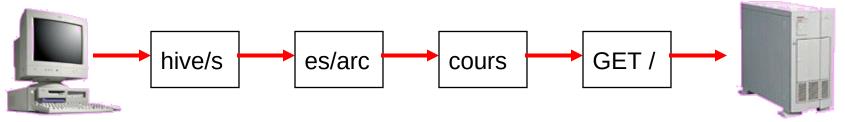


Request

Problem: Packet size

- Typical Web page is 10 kbytes
- On Ethernet, max IP packet is 1500 bytes

GET index.html

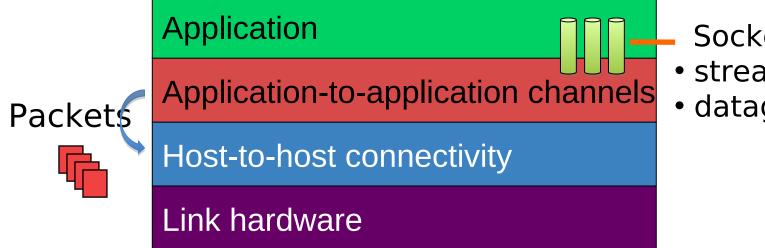


Solution: Split the data across multiple packets



Layering = Functional Abstraction

- Sub-divide the problem
 - Each layer relies on services from layer below
 - Each layer exports services to layer above
- Interface between layers defines interaction
 - Hides implementation details
 - Layers can change without disturbing other layers



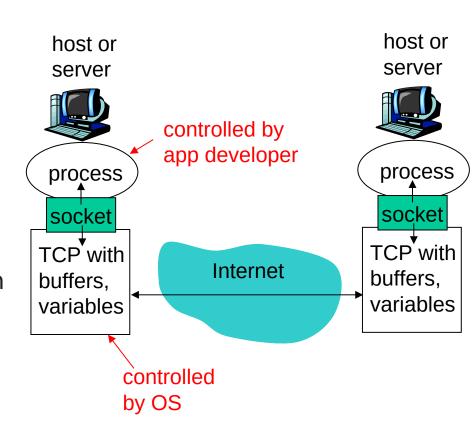
Sockets:

- streams TCP
- datagrams UDP



Sockets

- process sends/receives messages to/from its socket
- socket analogous to door
 - sending process shoves message out door
 - sending process relies on transport infrastructure on other side of door which brings message to socket at receiving process



API: (1) choice of transport protocol; (2) ability to fix a few parameters (more on this in next lecture!)

Source: Kurose & Ross (partial)

Internet transport protocols services

TCP service:

- connection-oriented: setup required between client and server processes
- reliable transport: between sending and receiving process
- flow control: sender won't overwhelm receiver
- congestion control: throttle sender when network overloaded
- does not provide: timing, minimum throughput guarantees, security

UDP service:

- unreliable data transfer between sending and receiving process
- does not provide: connection setup, reliability, flow control, congestion control, timing, throughput guarantee, or security

Q: why bother? Why is there a UDP?



UNIX Socket API

- Socket interface
 - Originally provided in Berkeley UNIX
 - Later adopted by all popular operating systems
 - Simplifies porting applications to different OSes
- In UNIX, everything is like a file
 - All input is like reading a file
 - All output is like writing a file
 - File is represented by an integer file descriptor
- API implemented as system calls
 - E.g., connect, read, write, close, ...

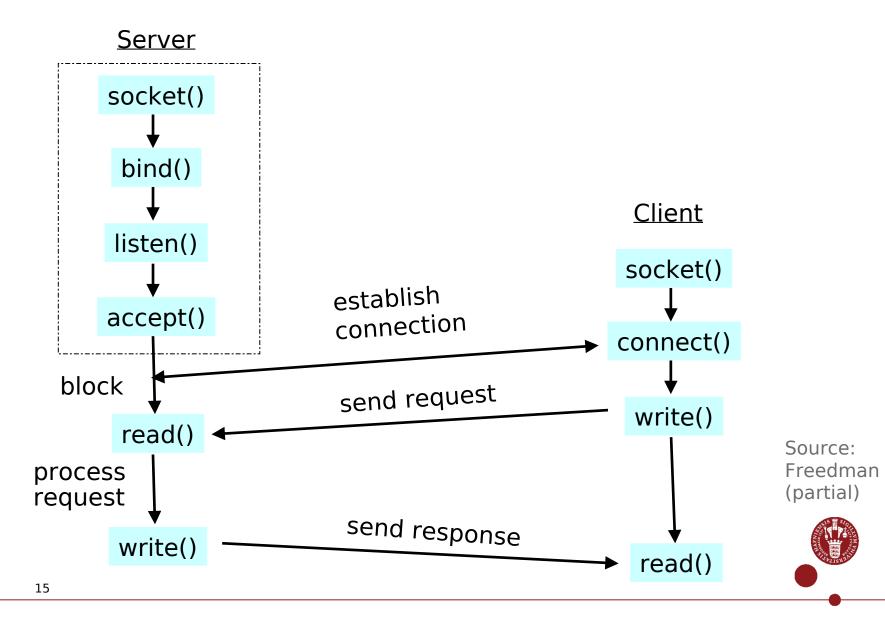


Identifying the Receiving Process

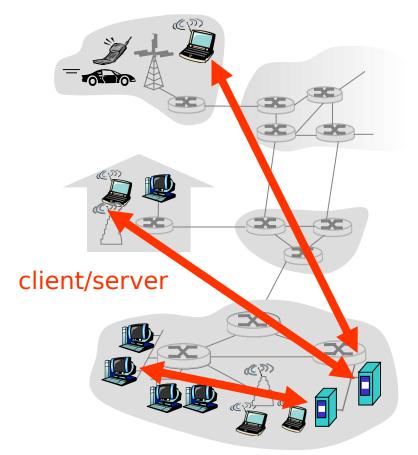
- Sending process must identify the receiver
 - The receiving end host machine
 - The specific socket in a process on that machine
- Receiving host
 - Destination address that uniquely identifies the host
 - Typically, high-level name translated to IP address (DNS)
 - For example, www.diku.dk \rightarrow 130.225.96.108
 - An IP address is a 32-bit quantity
- Receiving socket
 - Host may be running many different processes
 - Destination port that uniquely identifies the socket
 - A port number is a 16-bit quantity

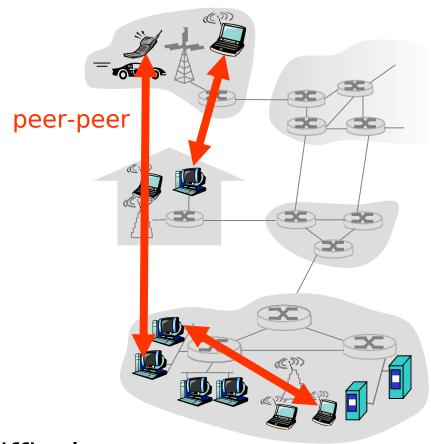


Client-Server TCP Sockets



Application Architectures



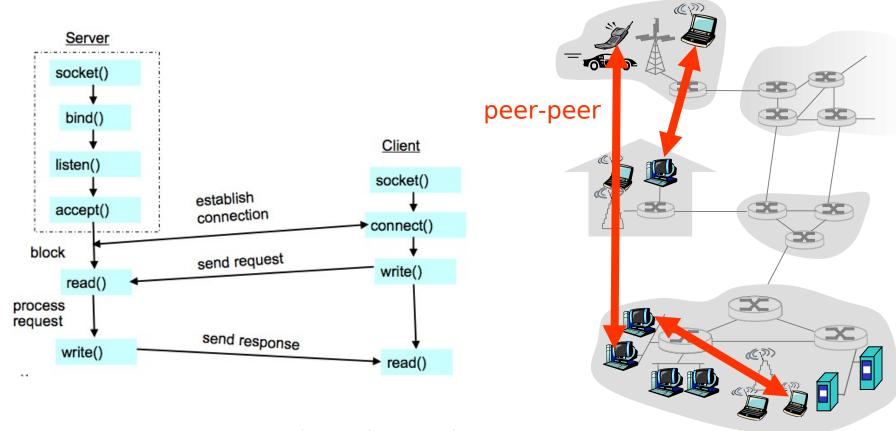


- •P2P highly scalable, but difficult to manage
- Hybrids also possible, e.g., Skype



Source: Kurose & Ross (partial)

Discussion: How do you set sockets up in a P2P application?



- •How many sockets in each peer?
- •What if many peers connect to one peer?



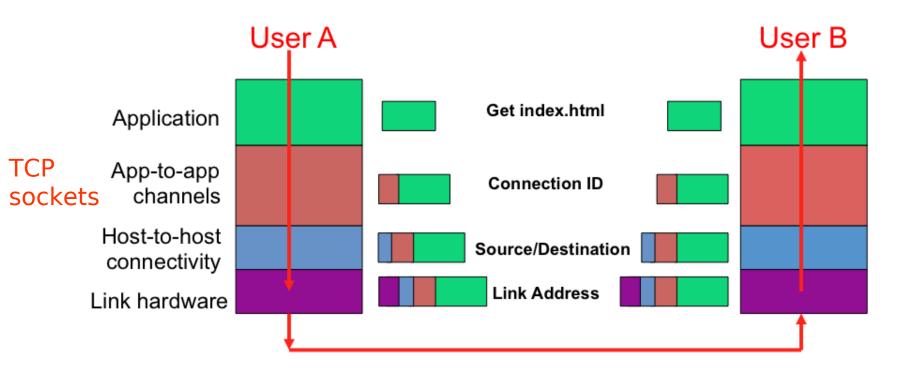
HTTP Basics

- HTTP layered over bidirectional byte stream
- Interaction
 - Client sends request to server, followed by response from server to client
 - Requests/responses are encoded in text
- Targets access to web objects
 - GET, POST, HEAD → HTTP/1.0
 - GET, POST, HEAD, PUT, DELETE → HTTP/1.1
- Stateless
 - Server maintains no info about past client requests
 - What about personalization? Data stored in back-end database; client sends "web cookie" used to lookup data



Layer Encapsulation in HTTP







HTTP Request Example

GET / HTTP/1.1

Host: sns.cs.princeton.edu

Accept: */*

Accept-Language: en-us

Accept-Encoding: gzip, deflate

User-Agent: Mozilla/5.0 (Macintosh; U; Intel Mac OS X

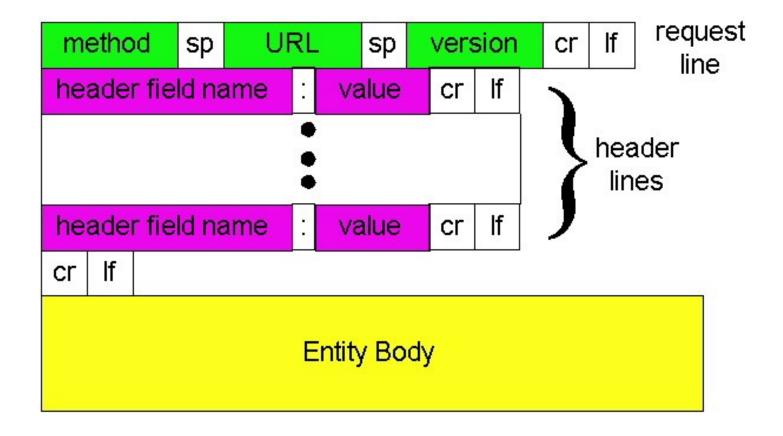
10.5; en-US; rv:1.9.2.13) Gecko/20101203

Firefox/3.6.13

Connection: Keep-Alive



HTTP Request





HTTP Response Example

HTTP/1.1 200 OK

Date: Wed, 02 Feb 2011 04:01:21 GMT

Server: Apache/2.2.3 (CentOS)

X-Pingback: http://sns.cs.princeton.edu/xmlrpc.php

Last-Modified: Wed, 01 Feb 2011 12:41:51 GMT

ETag: "7a11f-10ed-3a75ae4a"

Accept-Ranges: bytes Content-Length: 4333

Keep-Alive: timeout=15, max=100

Connection: Keep-Alive

```
<!DOCTYPE html PUBLIC "-//W3C//DTD XHTML 1.0 Transitional//EN" "http://www.w3.org/TR/xhtml1/DTD/xhtml1-transitional.dtd">
```

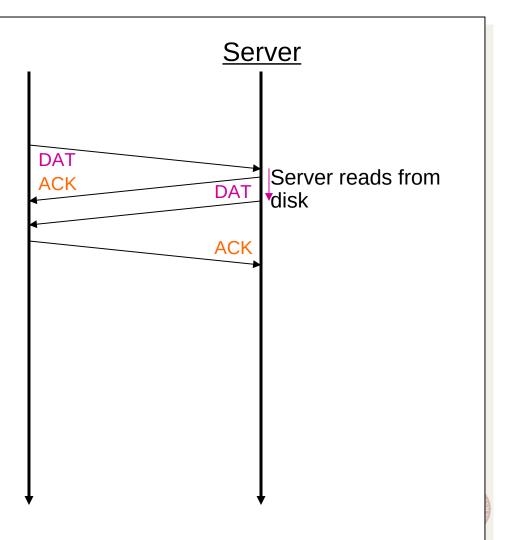



Single Transfer Example

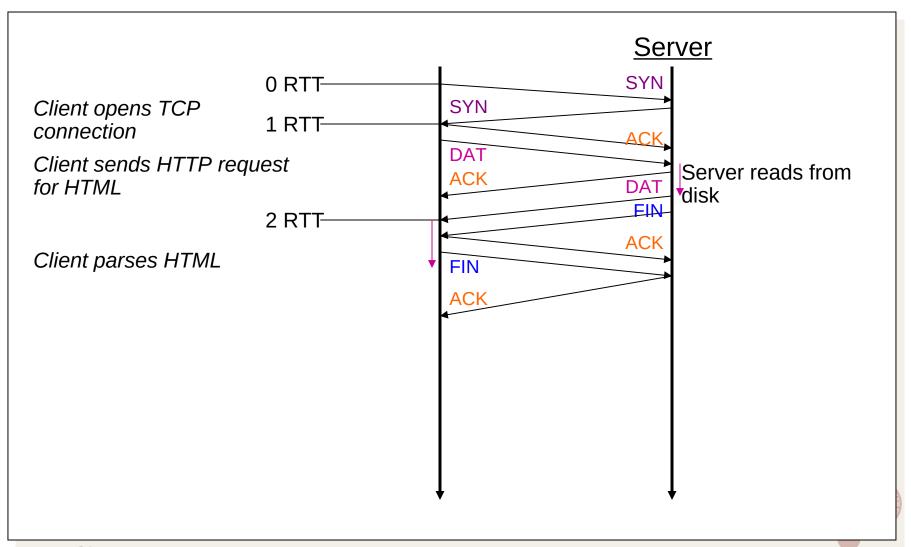
Source: Freedman

Client sends HTTP request for HTML

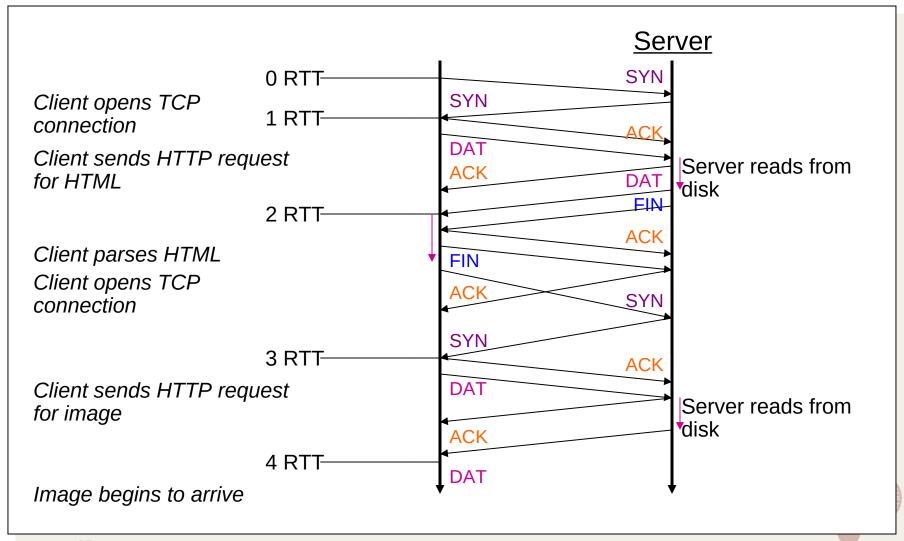
Client parses HTML



Single Transfer Example



Single Transfer Example

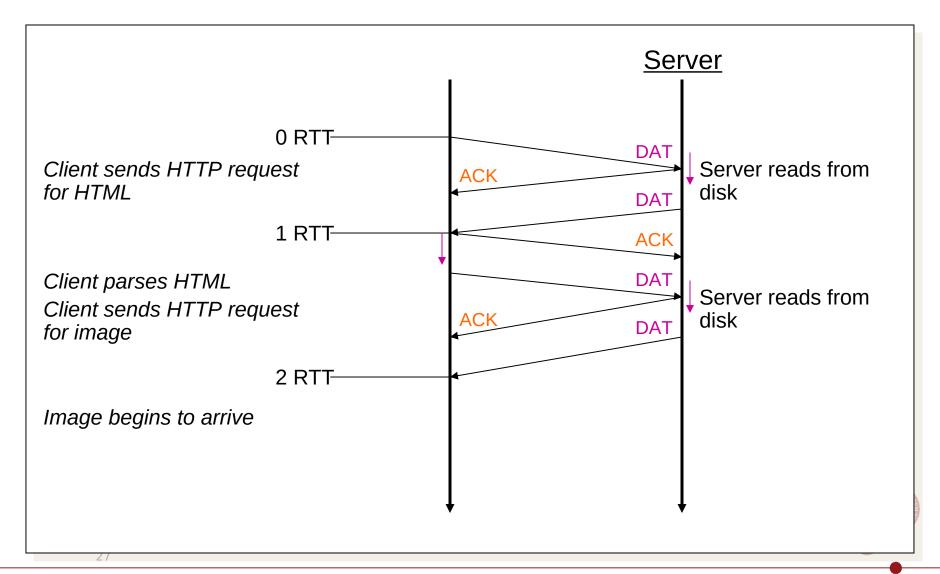


Problems with simple model

- Multiple connection setups
 - Three-way handshake each time (TCP "synchronizing" stream)
- Lots of extra connections
 - Increases server state/processing
 - Server forced to keep connection state
- Later we will see also that
 - Short transfers are hard on stream protocol (TCP)
 - How much data should it send at once?
 - Congestion avoidance: Takes a while to "ramp up" to high sending rate (TCP "slow start")
 - Loss recovery is poor when not "ramped up"



Persistent Connection Example



Persistent HTTP

Non-persistent HTTP issues:

- Requires 2 RTTs per object
- OS must allocate resources for each TCP connection
- But 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 are sent over connection



Persistent HTTP

<u>Persistent without</u> <u>pipelining:</u>

- Client issues new request only when previous response has been received
- One RTT for each object

Persistent with pipelining:

- Default in HTTP/1.1 spec
- Client sends requests as soon as it encounters referenced object
- As little as one RTT for all the referenced objects
- Server must handle responses in same order as requests
- •Persistent without pipelining most common: When does pipelining work best?
- •Multiple parallel requests or pipelined requests?



HTTP Caching

- Clients often cache documents
 - When should origin be checked for changes?
 - Every time? Every session? Date?
- HTTP includes caching information in headers
 - HTTP 0.9/1.0 used: "Expires: <date>"; "Pragma: no-cache"
 - HTTP/1.1 has "Cache-Control"
 - "No-Cache", "Private", "Max-age: <seconds>"
 - "E-tag: <opaque value>"
- If not expired, use cached copy
- If expired, use condition GET request to origin
 - "If-Modified-Since: <date>", "If-None-Match: <etag>"
 - 304 ("Not Modified") or 200 ("OK") response



HTTP Conditional Request

GET / HTTP/1.1

Host: sns.cs.princeton.edu

User-Agent: Mozilla/5.0 (Macintosh; U; Intel

Mac OS X 10.5; en-US; rv:1.9.2.13)

Connection: Keep-Alive

If-Modified-Since: Tue, 1 Feb 2011 17:54:18

GMT

If-None-Match: "7a11f-10ed-3a75ae4a"

HTTP/1.1 304 Not Modified

Date: Wed, 02 Feb 2011 04:01:21

GMT

Server: Apache/2.2.3 (CentOS)

ETag: "7a11f-10ed-3a75ae4a"

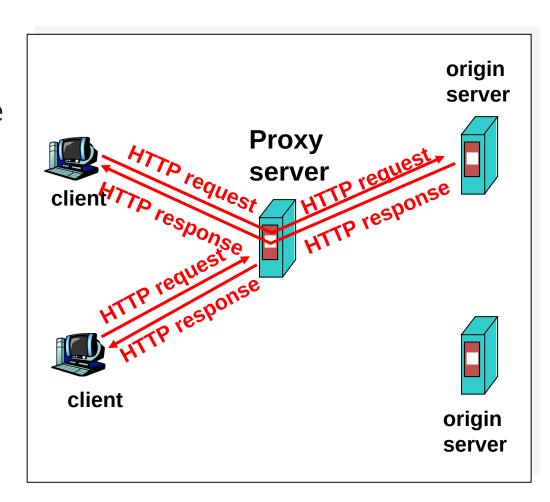
Accept-Ranges: bytes

Keep-Alive: timeout=15, max=100

Connection: Keep-Alive

Web Proxy Caches

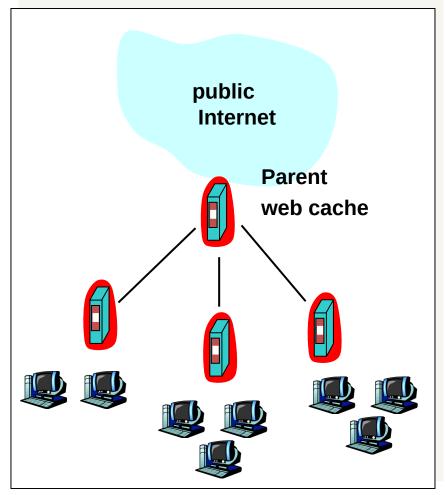
- User configures browser: Web accesses via cache
- Browser sends all HTTP requests to cache
 - Object in cache: cache returns object
 - Else: cache requests object from origin, then returns to client





When a single cache isn't enough

- What if the working set is > proxy disk?
 - Cooperation!
- A static hierarchy
 - Check local
 - If miss, check siblings
 - If miss, fetch through parent
- Internet Cache Protocol (ICP)
 - ICPv2 in RFC 2186 (& 2187)
 - UDP-based, short timeout





Streaming multimedia: DASH

- DASH: Dynamic, Adaptive Streaming over HTTP
- server:
 - divides video file into multiple chunks
 - each chunk stored, encoded at different rates
 - manifest file: provides URLs for different chunks
- client:
 - periodically measures server-to-client bandwidth
 - consulting manifest, requests one chunk at a time
 - chooses maximum coding rate sustainable given current bandwidth
 - can choose different coding rates at different points in time (depending on available bandwidth at time)

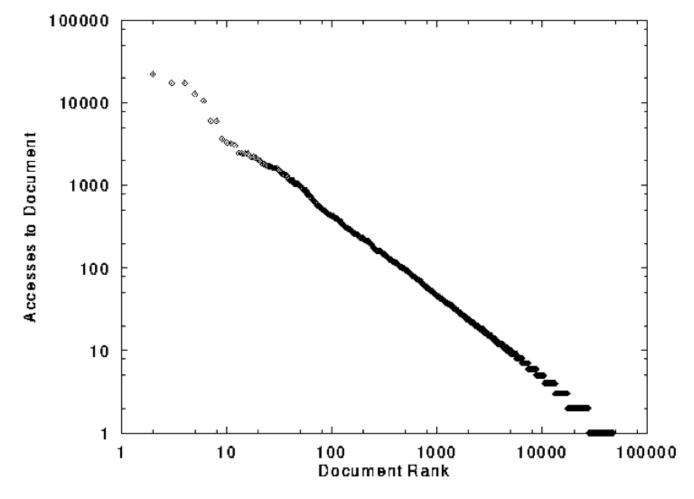


Streaming multimedia: DASH

- DASH: Dynamic, Adaptive Streaming over HTTP
- "intelligence" at client: client determines
 - when to request chunk (so that buffer starvation, or overflow does not occur)
 - what encoding rate to request (higher quality when more bandwidth available)
 - where to request chunk (can request from URL server that is "close" to client or has high available bandwidth)



Web traffic has cacheable workload





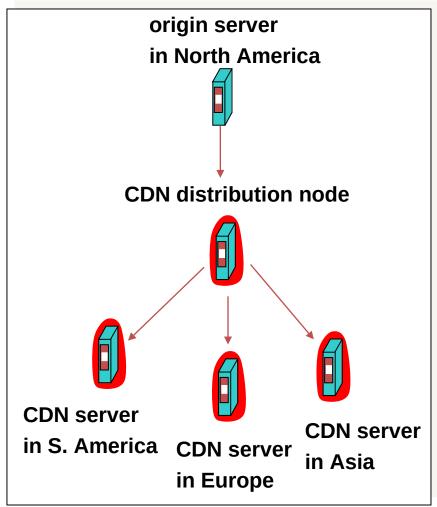


Content Distribution Networks (CDNs)

 Content providers are CDN customers

Content replication

- CDN company installs thousands of servers throughout Internet
 - In large datacenters
 - Or, close to users
- CDN replicates customers' content
- When provider updates content, CDN updates servers





Content Distribution Networks & Server Selection

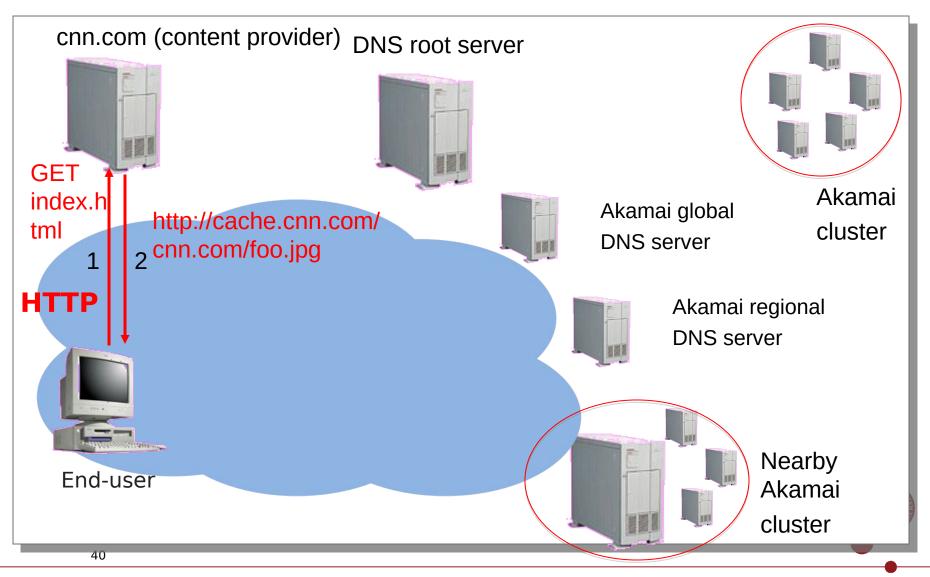
- Replicate content on many servers
- Challenges
 - How to replicate content
 - Where to replicate content
 - How to find replicated content
 - How to choose among known replicas
 - How to direct clients towards replica

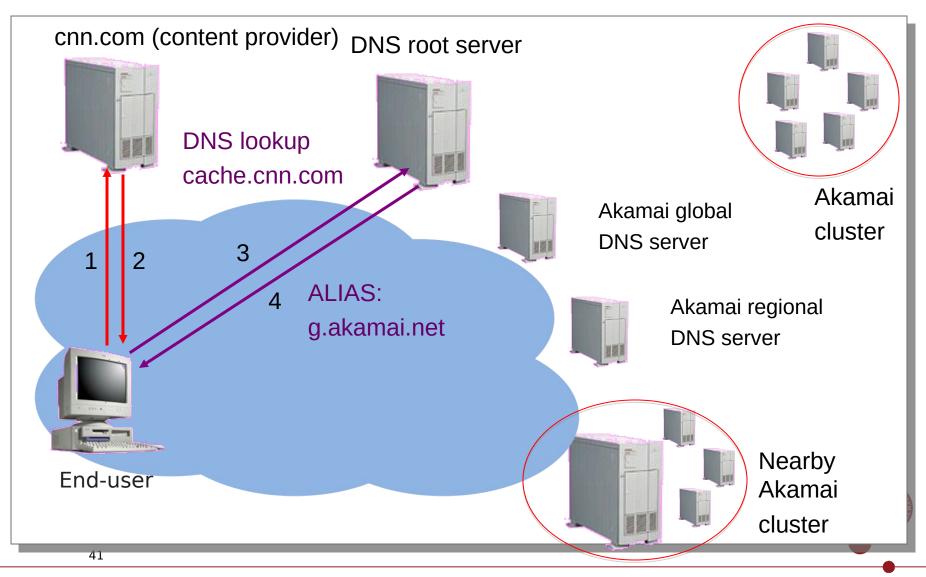


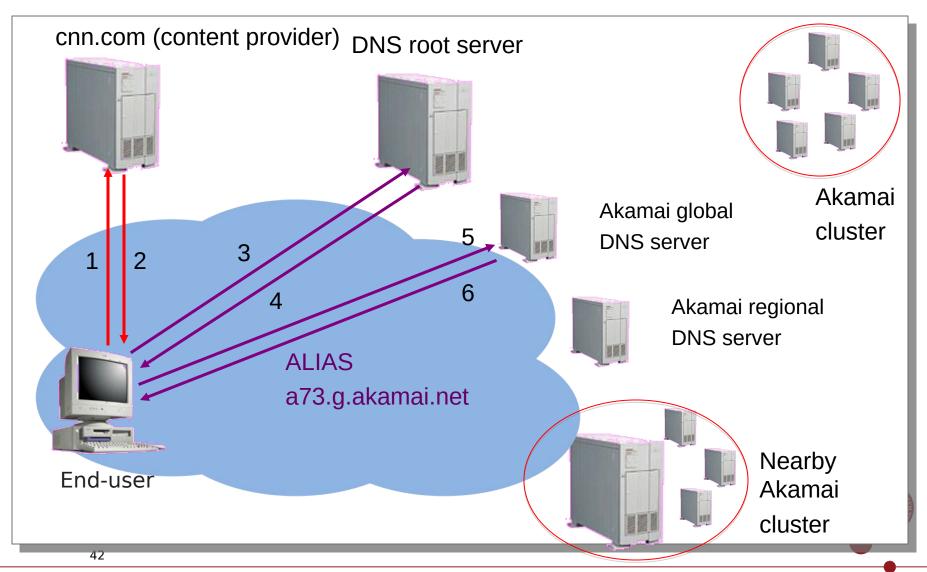
Server Selection

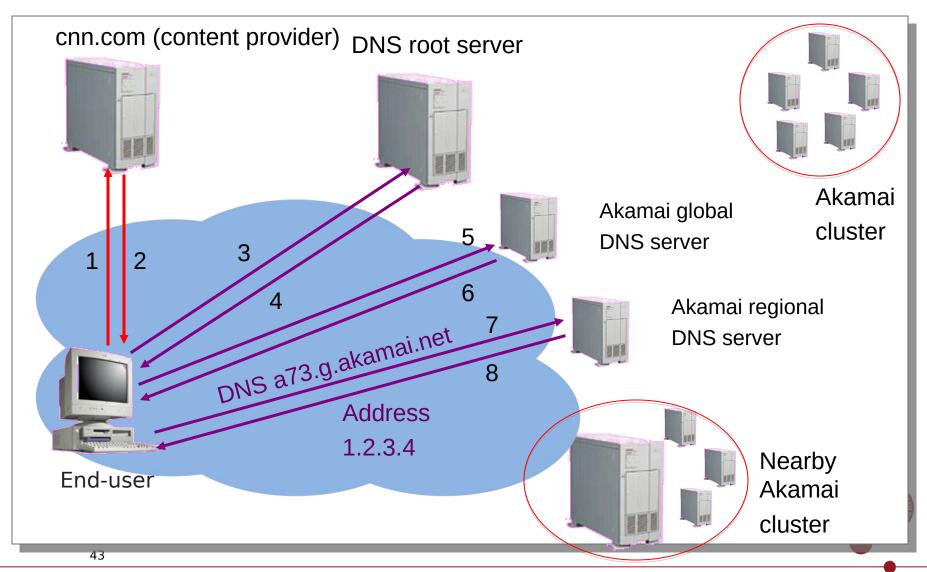
- Which server?
 - Lowest load: to balance load on servers
 - Best performance: to improve client performance
 - Based on Geography? RTT? Throughput? Load?
 - Any alive node: to provide fault tolerance
- How to direct clients to a particular server?
 - As part of routing: anycast, cluster load balancing
 - As part of application: HTTP redirect
 - As part of naming: DNS
 - We will explain some of these techniques better later in the course!

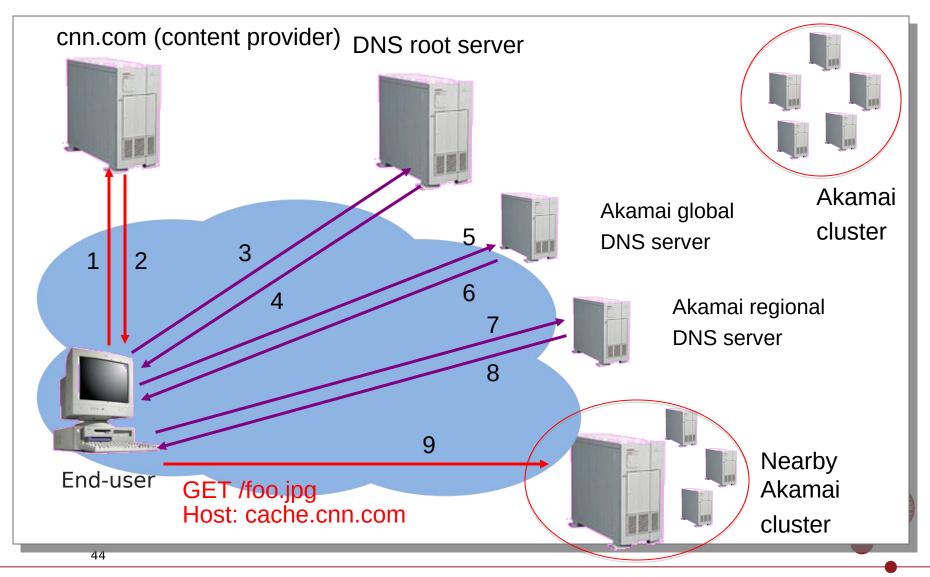


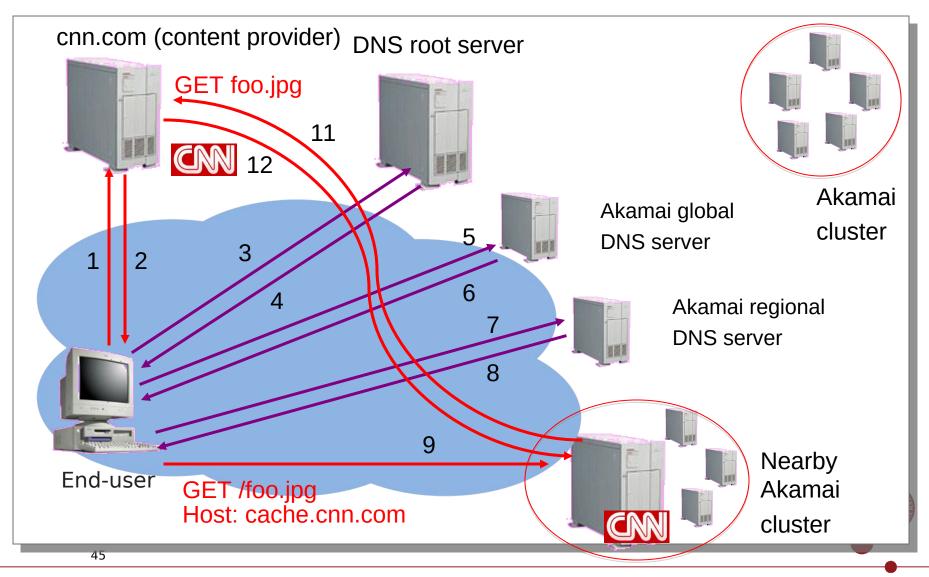


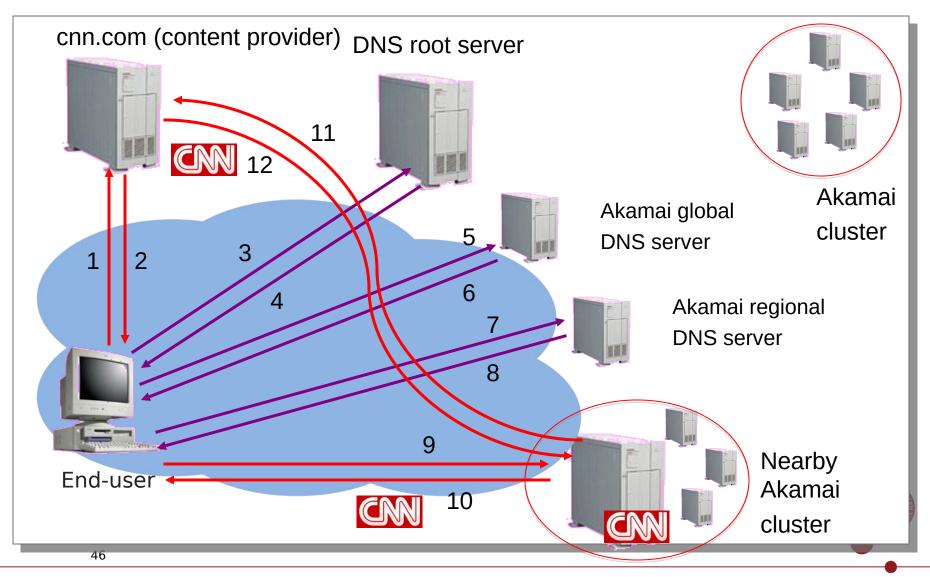












Summary

- Network applications
 - Email, Web → more in textbook, Chapter 2!
- Socket abstraction
 - Communication between processes
 - Client / Server, Peer-to-Peer
- HTTP concepts
 - Web objects, request / response (pull)
 - Persistent connections, web proxies
 - · Caching, content delivery

