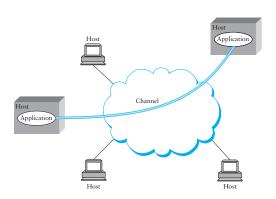
Outline

- Networking overview
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- Setwork file systems

Computer networking



- Goal: two applications on different computers exchange data
- Requires inter-process (not just inter-node) communication

The 7-Layer and 4-Layer Models

	OSI	TCP/IP
7	Application	Applications (FTP, SMTP, HTTP, etc.)
6	Presentation	
5	Session	
4	Transport	TCP (host-to-host)
3	Network	IP
2	Data link	Network access (usually Ethernet)
1	Physical	

Physical Layer

Computers send bits over physical links

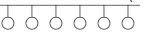
- E.g., Coax, twisted pair, fiber, radio, ...
- Bits may be encoded as multiple lower-level "chips"

Two categories of physical links

- Point-to-point networks (e.g., fiber, twisted pair):



- Shared transmission medium networks (e.g., coax, radio):



- Any message can be seen by all nodes
- Allows broadcast/multicast, but introduces contention

One important constraint: speed of light

- $\sim 300,000~\mathrm{km/sec}$ in a vacum, slower in fiber

SF
$$\xrightarrow{\geq \sim 15 \text{ msec}}$$
 NYC Moore's law does not apply!

Link Layer, Indirect Connectivity

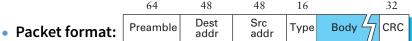
- When no direct physical connection to destination
- Hop through multiple devices



- Allows links and devices to be shared for multiple purposes
- Must determine which bits are part of which messages intended for which destinations
- Packet switched networks
 - Pack a bunch of bytes together intended for same destination
 - Slap a *header* on packet describing where it should go

Link Layer: Ethernet

- Originally designed for shared medium (coax), now generally not shared medium (switched)
- Vendors give each device a unique 48-bit MAC address
 - Specifies which card should receive a packet
- Ethernet switches can scale to switch local area networks. (thousands of hosts), but not much larger



- - Preamble helps device recognize start of packet
 - CRC allows receiving card to ignore corrupted packets
 - Body up to 1,500 bytes for same destination
 - All other fields must be set by sender's OS (NIC cards tell the OS what the card's MAC address is, Special addresses used for broadcast/multicast)

Network Layer: Internet Protocol (IP)

IP used to connect multiple networks

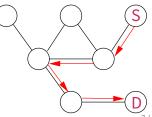
- Runs over a variety of physical networks
- Hence can connect Ethernet, DSL, mobile networks, etc.
- Most computers today speak IP

Every host has a unique 4-byte IP address (16-bytes for IPv6)

- (Or at least thinks it has, when there is address shortage)
- E.g., www.ietf.org \rightarrow 104.20.0.85

Packets are routed based on destination IP address

- Address space is structured to make routing practical at global scale
- E.g., 171.66.*.* goes to Stanford
- So packets need IP addresses in addition to MAC addresses



UDP and TCP

UDP and TCP most popular protocols on IP

- Both use 16-bit *port* number as well as 32-bit IP address
- Applications bind a port & receive traffic to that port

UDP – unreliable datagram protocol

- Exposes packet-switched nature of Internet
- Sent packets may be dropped, reordered, even duplicated (but generally not corrupted)

TCP – transmission control protocol

- Provides illusion of a reliable "pipe" between two processes on two different machines
- Masks lost & reordered packets so apps don't have to worry
- Handles congestion & flow control

Principles: Packet Switching & Layering

Packet switching

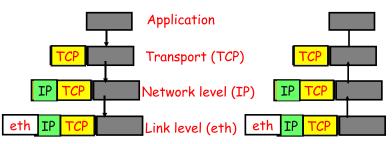
- A packet is a self contained unit of data which contains information necessary for it to reach its destination
- Independently, for each arriving packet, compute its outgoing link
- Makes forwarding simple (depends only on packet)

Layering

- Break system functionality into a hierarchy of layers
- Each layer uses only the service of the layer below it
- Layers communicate sequentially with the layers above or below

Principle: Encapsulation

- Stick packets inside packets
- How you realize packet switching and layering in a system
 - E.g., an Ethernet packet may encapsulate an IP packet
 - An IP router forwards a packet from one Ethernet to another, creating a new Ethernet packet containing the same IP packet
 - In principle, an inner layer should not depend on outer layers (not always true)



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Unreliability of IP

- Network does not deliver packets reliably
 - May drop packets, reorder packets, delay packets
 - May even corrupt packets, or duplicate them
- How to implement reliable TCP on top of IP network?
 - Note: This is entirely the job of the OS at the end nodes
- Straw man: Wait for ack for each packet
 - Send a packet, wait for acknowledgment, send next packet
 - If no ack, timeout and try again
- Problems?

Unreliability of IP

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 - Send a packet, wait for acknowledgment, send next packet
 - If no ack, timeout and try again
- Problems:
 - Low performance over high-delay network (bandwidth is one packet per round-trip time)
 - Possible congestive collapse of network (if everyone keeps retransmitting when network overloaded)

Performance: Bandwidth-delay

- Network delay over WAN will never improve much
- But throughput (bits/sec) is constantly improving
- Can view network as a pipe



- For full utilization want # bytes in flight > bandwidth × delay (But don't want to overload the network, either)
- What if protocol doesn't involve bulk transfer?
 - E.g., ping-pong protocol will have poor throughput
- Another implication: Concurrency & response time critical for good network utilization

A little bit about TCP

- Want to save network from congestion collapse
 - Packet loss usually means congestion, so back off exponentially
- Want multiple outstanding packets at a time
 - Get transmit rate up to *n*-packet window per round-trip
- Must figure out appropriate value of n for network
 - Slowly increase transmission by one packet per acked window
 - When a packet is lost, cut window size in half
- Connection set up and teardown complicated
 - Sender never knows when last packet might be lost
 - Must keep state around for a while after close
- Lots more hacks required for good performance
 - Initially ramp n up faster (but too fast caused collapse in 1986 [Jacobson], so TCP had to be changed)
 - Fast retransmit when single packet lost

Lots of OS issues for TCP

Have to track unacknowledged data

- Keep a copy around until recipient acknowledges it
- Keep timer around to retransmit if no ack
- Receiver must keep out of order segments & reassemble

When to wake process receiving data?

- E.g., sender calls write (fd, message, 8000);
- First TCP segment arrives, but is only 512 bytes
- Could wake recipient, but useless w/o full message
- TCP sets "PUSH" bit at end of 8000 byte write data

When to send short segment, vs. wait for more data

- Usually send only one unacked short segment
- But bad for some apps, so provide NODELAY option

Must ack received segments very quickly

- Otherwise, effectively increases RTT, decreasing bandwidth

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Sockets

- Abstraction for communication between machines
- Datagram sockets: Unreliable message delivery
 - With IP, gives you UDP
 - Send atomic messages, which may be reordered or lost
 - Special system calls to read/write: send/recv
- Stream sockets: Bi-directional pipes
 - With IP, gives you TCP
 - Bytes written on one end read on the other
 - Reads may not return full amount requested—must re-read

Socket naming

- TCP & UDP name communication endpoints by
 - E.g., 32-bit IPv4 address specifies machine (128 bits for IPv6)
 - 16-bit TCP/UDP port number demultiplexes within host
- A connection is thus named by 5 components
 - Protocol (TCP), local IP, local port, remote IP, remote port
 - TCP requires connected sockets, but not UDP
- OS keeps connection state in protocol control block (PCB) structure
 - Keep all PCB's in a hash table
 - When packet arrives (if destination IP address belongs to host), use
 5-tuple to find PCB and determine what to do with packet

System calls for using TCP

Server socket - make socket

bind – assign address listen – listen for clients

^{*}This call to bind is optional; connect can choose address & port.

Using UDP

- Call socket with SOCK_DGRAM, bind as before
- New system calls for sending individual packets

 - Must send/get peer address with each packet
- Can use UDP in connected mode
 - connect assigns remote address
 - send/recv syscalls, like sendto/recvfrom w/o last 2 args

Uses of connected UDP sockets

Kernel demultplexes packets based on port

- Allows different processes getting packets from different peers
- For security, ports < 1024 usually can't be bound
- But can safely inherit UDP port below that connected to one particular peer

Feedback based on ICMP messages

- Say no process has bound UDP port you sent packet to...
- With sendto, you might think network dropping packets
- Server sends port unreachable message, but only detect it when using connected sockets

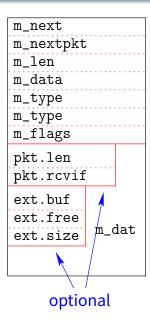
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Socket implementation

- Need to implement layering efficiently
 - Add UDP header to data, Add IP header to UDP packet, ...
 - De-encapsulate Ethernet packet so IP code doesn't get confused by Ethernet header
- Don't store packets in contiguous memory
 - Moving data to make room for new header would be slow
- BSD solution: mbufs [Leffler]
 (Note [Leffler] calls m_nextpkt by old name m_act)
 - Small, fixed-size (256 byte) structures
 - Makes allocation/deallocation easy (no fragmentation)
- BSD Mbufs working example for this lecture
 - Linux uses sk_buffs, which are similar idea

mbuf details



- Packets made up of multiple mbufs
 - Chained together by m_next
 - Such linked mbufs called *chains*
- Chains linked with m_nextpkt
 - Linked chains known as queues
 - E.g., device output queue
- Total mbuf size 256 B ⇒ ~230 data bytes (depends on size of pointers)
 - First in chain has pkt header
- Cluster mbufs have more data
 - ext header points to data
 - Up to 2 KB not collocated with mbuf
 - m_dat not used
- m_flags is bitwise or of various bits
 - E.g., if cluster, or if pkt header used

Adding/deleting data with mbufs

- m_data always points to start of data
 - Can be m_dat, or ext.buf for cluster mbuf
 - Or can point into middle of that area
- To strip off a packet header (e.g., TCP/IP)
 - Increment m_data, decrement m_len
- To strip off end of packet
 - Decrement m_len
- Can add data to mbuf if buffer not full
- Otherwise, add data to chain
 - Chain new mbuf at head/tail of existing chain

mbuf utility functions

- mbuf *m_copym(mbuf *m, int off, int len, int wait);
 - Creates a copy of a subset of an mbuf chain
 - Doesn't copy clusters, just increments reference count
 - wait says what to do if no memory (wait or return NULL)
- void m_adj(struct mbuf *mp, int len);
 - Trim |len| bytes from head or (if negative) tail of chain
- mbuf *m_pullup(struct mbuf *n, int len);
 - Put first len bytes of chain contiguously into first mbuf
- Example: Ethernet packet containing IP datagram
 - Trim Ethernet header using m_adj
 - Call m_pullup (n, sizeof (ip_hdr));
 - Access IP header as regular C data structure

Socket implementation

Each socket fd has associated socket structure with:

- Send and receive buffers
- Queues of incoming connections (on listen socket)
- A protocol control block (PCB)
- A protocol handle (struct protosw *)

PCB contains protocol-specific info. E.g., for TCP:

- Pointer to IP TCB with source/destination IP address and port
- Information about received packets & position in stream
- Information about unacknowledged sent packets
- Information about timeouts
- Information about connection state (setup/teardown)

protosw structure

Goal: abstract away differences between protocols

- In C++, might use virtual functions on a generic socket struct
- Here just put function pointers in protosw structure

Also includes a few data fields

- type, domain, protocol to match socket syscall args, so know which protosw to select
- flags to specify important properties of protocol

Some protocol flags:

- ATOMIC exchange atomic messages only (like UDP, not TCP)
- ADDR address given with messages (like unconnected UDP)
- CONNREQUIRED requires connection (like TCP)
- WANTRCVD notify socket of consumed data (e.g., so TCP can wake up a sending process blocked by flow control)

protosw functions

- pr_slowtimo called every 1/2 sec for timeout processing
- pr_drain called when system low on space
- pr_input takes mbuf chain of data to be read from socket
- pr_output takes mbuf chain of data written to socket
- pr_usrreq multi-purpose user-request hook
 - Used for bind/listen/accept/connect/disconnect operations
 - Used for out-of-band data

Network interface cards

Each NIC driver provides an ifnet data structure

- Like protosw, tries to abstract away the details

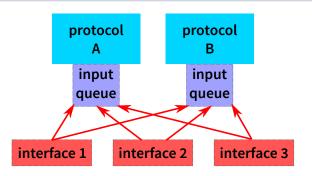
Data fields:

- Interface name (e.g., "eth0")
- Address list (e.g., Ethernet address, broadcast address, ...)
- Maximum packet size
- Send queue

Function pointers

- if_output prepend header and enqueue packet
- if_start start transmitting queued packets
- Also ioctl, timeout, initialize, reset

Input handling



- NIC driver figures out protocol of incoming packet
- Enqueues packet for appropriate protocol handler
 - If queue full, drop packet (can create livelock [Mogul])
- Posts "soft interrupt" for protocol-layer processing
 - Runs at lower priority than hardware (NIC) interrupt
 ...but higher priority than process-context kernel code

Routing

An OS must route all transmitted packets

- Machine may have multiple NICs plus "loopback" interface
- Which interface should a packet be sent to, and what MAC address should packet have?

Routing is based purely on the destination address

- Even if host has multiple NICs w. different IP addresses
- (Though OSes have features to redirect based on source IP)

OS maintains routing table

Maps IP address & prefix-length → next hop

Use radix tree for efficient lookup

- Branch at each node in tree based on single bit of target
- When you reach leaf, that is your next hop

Most OSes provide packet forwarding

Received packets for non-local address routed out another interface

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Network file systems

• What's a network file system?

- Looks like a file system (e.g., FFS) to applications
- But data potentially stored on another machine
- Reads and writes must go over the network
- Also called distributed file systems

Advantages of network file systems

- Easy to share if files available on multiple machines
- Often easier to administer servers than clients
- Access way more data than fits on your local disk
- Network + remote buffer cache faster than local disk

Disadvantages

- Network + remote disk slower than local disk
- Network or server may fail even when client OK
- Complexity, security issues

NFS version 2 [Sandberg]

- Background: ND (networked disk)
 - Creates disk-like device even on diskless workstations
 - Can create a regular (e.g., FFS) file system on it
 - But no sharing—Why?
- ND idea still used today by Linux NBD
 - Useful for network booting/diskless machines, not file sharing
- Some Goals of NFS
 - Access same FS from multiple machines simultaneously
 - Maintain Unix semantics
 - Crash recovery
 - Competitive performance with ND
- NFS version 2 protocol specified in [RFC 1094]

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 - Can create a regular (e.g., FFS) file system on it
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 - FFS assumes disk doesn't change under it
- ND idea still used today by Linux NBD
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 - Crash recovery
 - Competitive performance with ND
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NFS implementation

- Virtualized the file system with vnodes
 - Basically poor man's C++ (like protosw struct)
- Vnode structure represents an open (or openable) file
- Bunch of generic "vnode operations":
 - lookup, create, open, close, getattr, setattr, read, write, fsync, remove, link, rename, mkdir, rmdir, symlink, readdir, readlink, ...
 - Called through function pointers, so most system calls don't care what type of file system a file resides on
- NFS vnode operations perform Remote Procedure Calls (RPC)
 - Client sends request to server over network, awaits response
 - Each system call may require a series of RPCs
 - System mostly determined by RPC [RFC 1831] Protocol
 - Uses XDR protocol specification language [RFC 1832]

Stateless operation

- Designed for "stateless operation"
 - Motivated by need to recover from server crashes
- Requests are self-contained
- Requests are idempotent
 - Unreliable UDP transport
 - Client retransmits requests until it gets a reply
 - Writes must be stable before server returns
- Can this really work?

Stateless operation

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mostly

- Requests are \(idempotent \)
 - Unreliable UDP transport
 - Client retransmits requests until it gets a reply
 - Writes must be stable before server returns
- Can this really work?
 - Of course, FS not stateless it stores files
 - E.g., mkdir can't be idempotent second time dir exists
 - But many operations, e.g., read, write are idempotent

NFS version 3

- Same general architecture as NFS 2
- Specified in RFC 1813 (subset of Open Group spec)
 - XDR defines C structures that can be sent over network; includes tagged unions (to know which union field active)
 - Protocol defined as a set of Remote Procedure Calls (RPCs)
- New access RPC
 - Supports clients and servers with different uids/gids
- Better support for caching
 - Unstable writes while data still cached at client
 - More information for cache consistency
- Better support for exclusive file creation

NFSv3 File handles

```
struct nfs_fh3 {
  /* XDR notation for variable-length array
  * with 0-64 opaque bytes: */
  opaque data<64>;
};
```

- Server assigns an opaque file handle to each file
 - Client obtains first file handle out-of-band (mount protocol)
 - File handle hard to guess security enforced at mount time
 - Subsequent file handles obtained through lookups
- File handle internally specifies file system & file
 - Device number, i-number, *generation number*, ...
 - Generation number changes when inode recycled
- Handle generally doesn't contain filename
 - Clients may keep accessing an open file after it's renamed

File attributes

```
struct fattr3 {
  ftype3 type;
  uint32 mode;
  uint32 nlink;
  uint32 uid;
  uint32 gid;
  uint32 gid;
  uint64 size;
  uint64 used;
specdata3 rdev;
  uint64 fsid;
  uint64 fileid;
  nfstime3 atime;
  nfstime3 mtime;
  infstime3 ctime;
  infstime4 ctime;
};
```

- Most operations can optionally return fattr3
- Attributes used for cache-consistency

Lookup

```
struct diropargs3 {
                         struct lookup3resok {
 nfs_fh3 dir;
                           nfs_fh3 object;
 filename3 name;
                           post_op_attr obj_attributes;
};
                           post_op_attr dir_attributes;
                         };
union lookup3res switch (nfsstat3 status) {
case NFS3 OK:
 lookup3resok resok;
default:
 post_op_attr resfail;
};
```

- Maps $\langle \text{directory handle}, \text{filename} \rangle \rightarrow \text{handle}$
 - Client walks hierarchy one file at a time
 - No symlinks or file system boundaries crossed
 - Client must expand symlinks

Create

```
struct create3args {
 diropargs3 where;
 createhow3 how;
};
union createhow3 switch (createmode3 mode) {
case UNCHECKED:
case GUARDED:
 sattr3 obj_attributes;
case EXCLUSIVE:
 createverf3 verf;
};
```

- UNCHECKED succeed if file exists
- GUARDED fail if file exists
- EXCLUSIVE persistent record of create

Read

```
struct read3args { struct read3resok {
 nfs_fh3 file;
                       post_op_attr file_attributes;
 uint64 offset;
                       uint32 count;
 uint32 count;
                       bool eof;
};
                       opaque data<>;
union read3res switch (nfsstat3 status) {
case NFS3_OK:
 read3resok resok;
default:
 post_op_attr resfail;
};
```

- Offset explicitly specified (not implicit in handle)
- Client can cache result

Data caching

- Client can cache blocks of data read and written
- Consistency based on times in fattr3
 - mtime: Time of last modification to file
 - ctime: Time of last change to inode (Changed by explicitly setting mtime, increasing size of file, changing permissions, etc.)
- Algorithm: If mtime or ctime changed by another client, flush cached file blocks

Write discussion

- When is it okay to lose data after a crash?
 - Local file system?

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- When is it okay to lose data after a crash?
 - Local file system?
 If no calls to fsync, OK to lose 30 seconds of work after crash
 - Network file system?

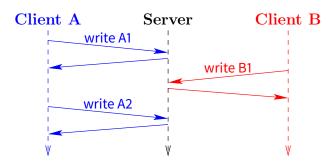
Write discussion

- When is it okay to lose data after a crash?
 - Local file system?
 If no calls to fsync, OK to lose 30 seconds of work after crash
 - Network file system?
 What if server crashes but not client?
 Application not killed, so shouldn't lose previous writes
- NFSv2 addresses problem by having server write data to disk before replying to a write RPC
 - Caused performance problems
- Could NFS2 clients just perform write-behind?
 - Implementation issues used blocking kernel threads on write
 - Semantics how to guarantee consistency after server crash
 - Solution: small # of pending write RPCs, but write through on close; if server crashes, client keeps re-writing until acked

NFSv2 write call

- On successful write, returns new file attributes
- Can NFSv2 keep cached copy of file after writing it?

Write race condition



Suppose client overwrites 2-block file

- Client A knows attributes of file after writes A1 & A2
- But client B could overwrite block 1 between the A1 & A2
- No way for client A to know this hasn't happened
- Must flush cache before next file read (or at least open)

NFSv3 Write arguments

```
struct write3args {
    nfs_fh3 file;
    uint64 offset;
    uint32 count;
    stable_how stable;
    opaque data<>;
};
enum stable_how {
    UNSTABLE = 0,
    DATA_SYNC = 1,
    FILE_SYNC = 2
    stable_how stable;
    opaque data<>;
};
```

Two goals for NFSv3 write:

- Don't force clients to flush cache after writes
- Don't equate *cache* consistency with *crash* consistency l.e., don't wait for disk just so another client can see data

Write results

```
struct write3resok {
                                     struct wcc_attr {
  wcc_data file_wcc;
                                       uint64 size;
  uint32 count;
                                       nfstime3 mtime;
  stable_how committed;
                                       nfstime3 ctime;
  writeverf3 verf;
                                     };
};
                                     struct wcc_data {
union write3res
                                       wcc attr *before:
    switch (nfsstat3 status) {
                                       post_op_attr after;
case NFS3_OK:
                                     };
  write3resok resok;
default:
  wcc_data resfail;
};
```

Several fields added to achieve these goals

Data caching after a write

- Write will change mtime/ctime of a file
 - "after" will contain new times
 - With NFSv2, would require cache to be flushed
- With NFSv3, "before" contains previous values
 - If before matches cached values, no other client has changed file
 - Okay to update attributes without flushing data cache

Write stability

- Server write must be at least as stable as requested
- If server returns write UNSTABLE
 - Means permissions okay, enough free disk space, ...
 - But data not on disk and might disappear (after crash)
- If DATA_SYNC, data on disk, maybe not attributes
- If FILE_SYNC, operation complete and stable

Commit operation

- Client cannot discard any UNSTABLE write
 - If server crashes, data will be lost
- COMMIT RPC commits a range of a file to disk
 - Invoked by client when client cleaning buffer cache
 - Invoked by client when user closes/flushes a file
- How does client know if server crashed?
 - Write and commit return writeverf3
 - Value changes after each server crash (can be boot time)
 - Client must resend all writes if verf value changes

Attribute caching

- Close-to-open consistency
 - Annoying if writes not visible after a file close (Edit file, compile on another machine, get old version)
 - Nowadays, all NFS opens fetch attributes from server
- Still, lots of other need for attributes (e.g., ls -al)
- Attributes cached between 5 and 60 seconds
 - Files recently changed more likely to change again
 - Do weighted cache expiration based on age of file
- Drawbacks:
 - Must pay for round-trip to server on every file open
 - Can get stale info when statting a file