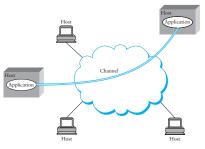
#### Outline

- Networking overview
- 2 Systems issues
- 3 OS networking facilities
- 4 Implementing networking in the kernel
- 5 Network file systems



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

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#### The 7-Layer and 4-Layer Models

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

#### **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 m km/sec in a vacum, slower in fiber

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

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#### Link Layer, Indirect Connectivity

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

source switch destination

- 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

	64	48	48	16	32
• Packet format:	Preamble	Dest addr	Src addr	Type	Body CRC

- 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)



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## Computer networking

## **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

necessary for it to reach its destination

- Makes forwarding simple (depends only on packet)

Break system functionality into a hierarchy of layers
Each layer uses only the service of the layer below it

## **Principles: Packet Switching & Layering**

- A packet is a self contained unit of data which contains information

- Independently, for each arriving packet, compute its outgoing link

- Layers communicate sequentially with the layers above or below

## 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

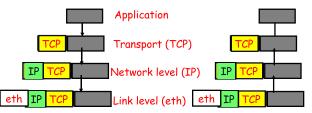
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#### **Principle: Encapsulation**

• 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?

- Networking overview

Packet switching

Layering

- 2 Systems issues
- OS networking facilities
- Implementing networking in the kernel
- 5 Network file systems

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Outline

## **Unreliability of IP**

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  - 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:
  - 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)

A little bit about TCP

## **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

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## Lots of OS issues for 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)

Outline

- Fast retransmit when single packet lost

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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

- Networking overview
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# Keep timer around to retransmit if no ack Receiver must keep out of order segments & reassemble

• When to wake process receiving data?

Have to track unacknowledged 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

- Keep a copy around until recipient acknowledges it

- 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

## 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

Client	Server
	socket – make socket
	bind – assign address
	listen – listen for clients
socket – make socket	
bind* – assign address	
connect - connect to lister	ning socket
	accept – accept connection

\*This call to bind is optional; connect can choose address & port.

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## **Uses of connected UDP sockets**

• Call socket with SOCK\_DGRAM, bind as before

#### New system calls for sending individual packets

- int sendto(int s, const void \*msg, int len, int flags, const struct sockaddr \*to, socklen\_t tolen);

**Using UDP** 

- int recvfrom(int s, void \*buf, int len, int flags, struct sockaddr \*from, socklen\_t \*fromlen);
- 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

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

## Outline **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

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- Networking overview
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## 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  $\Rightarrow$   $\sim$ 230 data bytes (depends on size of pointers)
    - First in chain has pkt header
  - Cluster mbufs have more data
    - ext header points to data

mbuf utility functions

• mbuf \*m\_copym(mbuf \*m, int off, int len, int wait);

- Doesn't copy clusters, just increments reference count

- Trim |len| bytes from head or (if negative) tail of chain

- wait says what to do if no memory (wait or return NULL)

- Creates a copy of a subset of an mbuf chain

• void m\_adj(struct mbuf \*mp, int len);

• mbuf \*m\_pullup(struct mbuf \*n, int len);

- 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

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#### **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)

- pr\_slowtimo called every 1/2 sec for timeout processing
- pr\_drain called when system low on space
- pr\_input returns mbuf chain of data read from socket
- pr\_output takes mbuf chain of data written to socket
- pr\_usrreg multi-purpose user-request hook
  - Used for bind/listen/accept/connect/disconnect operations
  - Used for out-of-band data

m\_nextpkt m\_len m\_data m\_type m\_type m\_flags pkt.len pkt.rcvif ext.buf ext.free m\_dat ext.size optional

m\_next

## protosw functions

- Put first len bytes of chain contiguously into first mbuf

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- Example: Ethernet packet containing IP datagram
  - Trim Ethernet header using m\_adj
  - Callm\_pullup (n, sizeof (ip\_hdr));
  - Access IP header as regular C data structure

## **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, ...)

Routing

- Machine may have multiple NICs plus "loopback" interface

Routing is based purely on the destination address

- Maps IP address & prefix-length  $\rightarrow$  next hop

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

- Branch at each node in tree based on single bit of target

Which interface should a packet be sent to, and what MAC address

- Maximum packet size
- Send queue

#### Function pointers

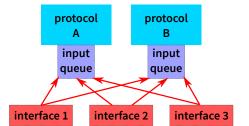
- if\_output prepend header and enqueue packet
- if\_start start transmitting queued packets
- Also ioctl, timeout, initialize, reset

An OS must route all transmitted packets

should packet have?

OS maintains routing table

## **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

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- When you reach leaf, that is your next hop

Use radix tree for efficient lookup

- Most OSes provide packet forwarding
  - Received packets for non-local address routed out another interface

#### 32/53

#### 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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- Background: ND (networked disk)
  - Creates disk-like device even on diskless workstations
  - 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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  - Access same FS from multiple machines simultaneously
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  - 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]

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#### **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?

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- 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

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#### **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

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File attributes	Lookup
<pre>struct f f f f f s s s s s s s s s s s s s s</pre>	<pre>struct diropargs3 {</pre>
Create	Read
<pre>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   GUADDED foil if file exists</pre>	<pre>struct read3args { struct read3resok {     nfs_fh3 file;    post_op_attr file_attributes;     uint64 offset;    uint32 count;     uint32 count;    bool eof; };</pre>
<ul> <li>GUARDED – fail if file exists</li> <li>EXCLUSIVE – persistent record of create</li> </ul>	<ul> <li>Offset explicitly specified (not implicit in handle)</li> <li>Client can cache result</li> </ul>

## Data caching

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

- 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

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## 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?

## 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

Write race condition

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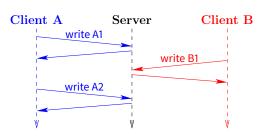
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## NFSv2 write call

<pre>struct writeargs {</pre>	union attrstat
fhandle file;	<pre>switch (stat status) {</pre>
unsigned beginoffset;	case NFS_OK:
unsigned offset;	fattr attributes;
unsigned totalcount;	default:
nfsdata data;	void;
};	};

attrstat NFSPROC\_WRITE(writeargs) = 8;

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



#### 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)

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s

}

}

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#### **NFSv3 Write arguments**

struct write3args { nfs\_fh3 file; uint64 offset; uint32 count; stable\_how stable; opaque data<>; };

#### enum stable\_how { UNSTABLE = 0, $DATA_SYNC = 1$ , $FILE_SYNC = 2$ };

#### • Two goals for NFSv3 write:

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

## Write results

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

#### Several fields added to achieve these goals

#### Data caching after a write

## Write stability

#### • 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

- 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

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## 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

#### Close-to-open consistency

- Annoying if writes not visible after a file close (Edit file, compile on another machine, get old version)

**Attribute caching** 

- 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

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