

## Networking

Don Porter → Vyas Sekar CSE 506

## Logical Diagram

Binary Formats Memory Allocators

Threads

System Calls

Today's Lecture

RCU

File System

Networking

Sync

Memory Management Device Drivers CPU Scheduler

Interrupts

Disk

Net

Consistency

Hardware

## Networking (2 parts)

#### ♦ Goals:

- \* Review networking basics
- ♦ Discuss APIs
- ♣ Trace how a packet gets from the network device to the application (and back)
- ♦ Understand Receive livelock and NAPI

## 4 to 7 layer diagram

(from Understanding Linux Network Internals)

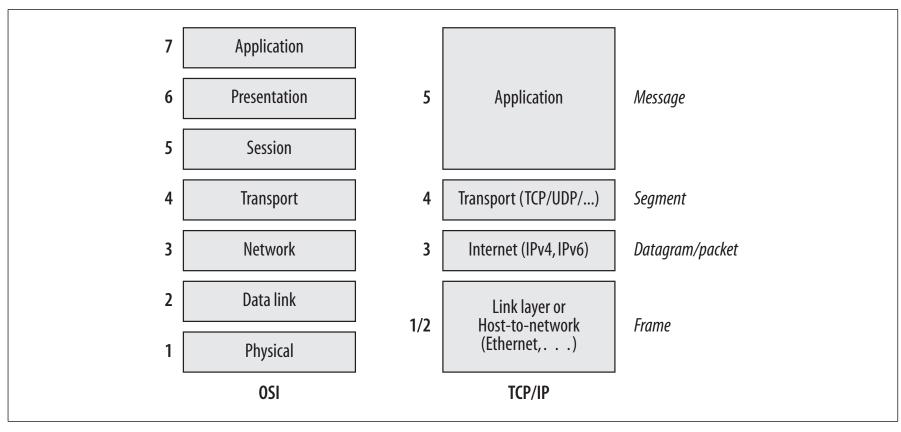


Figure 13-1. OSI and TCP/IP models

#### Nomenclature

♦ Frame: hardware

♦ Packet: IP

♦ Segment: TCP/UDP

♦ Message: Application

## TCP/IP Reality

- ♦ The OSI model is great for undergrad courses
- ♦ TCP/IP (or UDP) is what the majority of programs use
  - ♦ Some random things (like networked disks) just use ethernet + some custom protocols

## Ethernet (or 802.2 or 802.3)

- ♦ All slight variations on a theme (3 different standards)
- ♦ Simple packet layout:
  - ✦ Header: Type, source MAC address, destination MAC address, length, (and a few other fields)
  - ♦ Data block (payload)
  - ♦ Checksum
- ♦ Higher-level protocols "nested" inside payload
- † "Unreliable" no guarantee a packet will be delivered

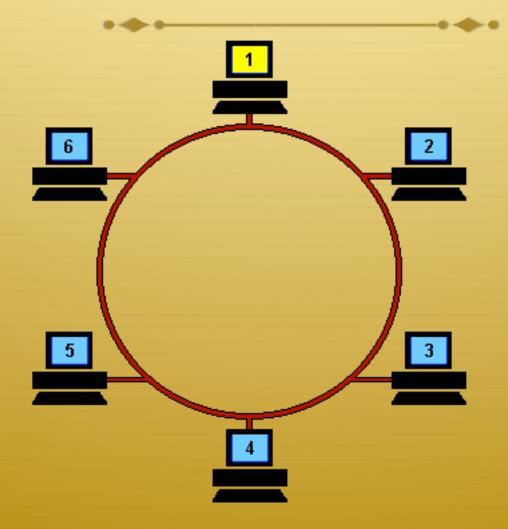
## Ethernet History

- ♦ Originally designed for a shared wire (e.g., coax cable)
- ♦ Each device listens to all traffic
  - → Hardware filters out traffic intended for other hosts
    - ♦ I.e., different destination MAC address
  - ♦ Can be put in "promiscuous" mode, and record everything (called a network sniffer)
- ♦ Sending: Device hardware automatically detects if another device is sending at same time
  - ♦ Random back-off and retry

## Early competition

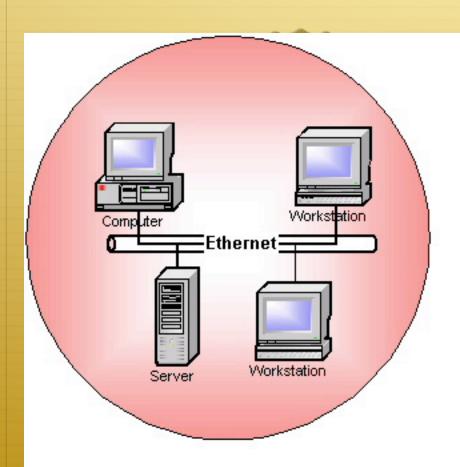
- ♦ Token-ring network: Devices passed a "token" around
  - → Device with the token could send; all others listened
  - ♦ Like the "talking stick" in a kindergarten class
- ♦ Send latencies increased proportionally to the number of hosts on the network
  - ♣ Even if they weren't sending anything (still have to pass the token)
- ♦ Ethernet has better latency under low contention and better throughput under high

## Token ring

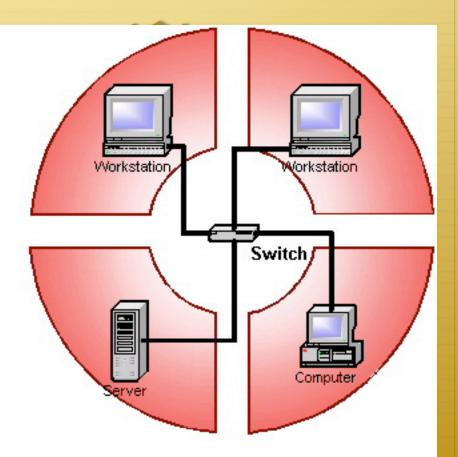


Source: http://www.datacottage.com/nch/troperation.htm

#### Shared vs Switched



**Shared Ethernet:**: 1 collision domain for multiple nodes. The possibility of collisions. Non-deterministic



**Switched Full Duplex Ethernet:** 1 collision domain per node. Use of switch. No possibility of collisions. Deterministic.

Source: http://www.industrialethernetu.com/courses/401\_3.htm

#### Switched networks

- ♦ Modern ethernets are switched
- ♦ What is a hub vs. a switch?
  - ♦ Both are a box that links multiple computers together
  - Hubs broadcast to all plugged-in computers (let computers filter traffic)
  - ♦ Switches track who is plugged in, only send to expected recipient
    - ♦ Makes sniffing harder ⊗

## Internet Protocol (IP)

- ♦ 2 flavors: Version 4 and 6
  - ♦ Version 4 widely used in practice---today's focus
- Provides a network-wide unique device address (IP address)
- ♦ This layer is responsible for routing data across multiple ethernet networks on the internet
  - ♦ Ethernet packet specifies its payload is IP
  - ↑ At each router, payload is copied into a new point-to-point ethernet frame and sent along

# Transmission Control Protocol (TCP)

- + Higher-level protocol that layers end-to-end reliability, transparent to applications
  - → Lots of packet acknowledgement messages, sequence numbers, automatic retry, etc.
  - Pretty complicated
- ♦ Applications on a host are assigned a *port* number
  - ♦ A simple integer from 0-64k
  - → Multiplexes many applications on one device
  - → Ports below 1k reserved for privileged applications

# User Datagram Protocol (UDP)

- ♦ The simple alternative to TCP
  - ♦ None of the frills (no reliability guarantees)
- ♦ Same port abstraction (1-64k)
  - ♦ But different ports
  - ♦ I.e., TCP port 22 isn't the same port as UDP port 22

## Some well-known ports

- ♦ 80 http
- ♦ 22 ssh

## Example

(from Understanding Linux Network Internals)

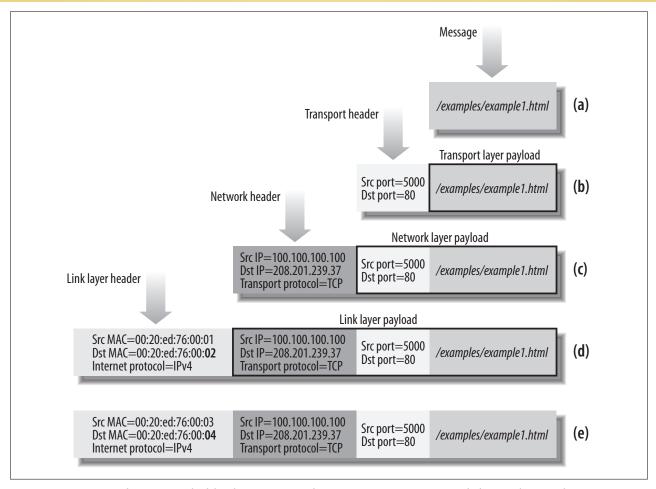


Figure 13-4. Headers compiled by layers: (a...d) on Host X as we travel down the stack; (e) on Router RT1

## Networking APIs

- ♦ Programmers rarely create ethernet frames
- ♦ Most applications use the **socket** abstraction
  - ♦ Stream of messages or bytes between two applications
  - \* Applications still specify: protocol (TCP vs. UDP), remote host address
    - ♦ Whether reads should return a stream of bytes or distinct messages
- ♦ While many low-level details are abstracted, programmers must understand basics of low-level protocols

## Sockets, cont.

- ♦ One application is the server, or listens on a predetermined port for new connections
- ♦ The client connects to the server to create a message channel
- ♦ The server accepts the connection, and they begin exchanging messages

#### Creation APIs

- † int socket(domain, type, protocol) create a file handle representing the communication endpoint
  - ♦ Domain is usually AF\_INET (IP4), many other choices
  - → Type can be STREAM, DGRAM, RAW
  - ♦ Protocol usually 0
- int bind(fd, addr, addrlen) bind this socket to a specific port, specified by addr
  - Can be INADDR\_ANY (don't care what port)

#### Server APIs

- int listen(fd, backlog) Indicate you want incoming connections
  - Backlog is how many pending connections to buffer until dropped
- † int accept(fd, addr, len, flags) Blocks until you get a
  connection, returns where from in addr
  - Return value is a new file descriptor for child
  - → If you don't like it, just close the new fd

#### Client APIs

- ♦ Both client and server create endpoints using socket()
  - ♦ Server uses bind, listen, accept
  - ♦ Client uses connect(fd, addr, addrlen) to connect to server
- ♦ Once a connection is established:
  - ♦ Both use send/recv
  - Pretty self-explanatory calls

## Client/server toy example

♦ Quick demo ...

♦ Client/server code from

http://www.linuxhowtos.org/C\_C++/socket.htm

## Linux implementation

- ♦ Sockets implemented in the kernel
  - ♦ So are TCP, UDP and IP
- ♦ Benefits:
  - Application doesn't need to be scheduled for TCP ACKs, retransmit, etc.
  - ♦ Kernel trusted with correct delivery of packets
- ♦ A single system call (i386):
  - \* sys\_socketcall(call, args)
    - ✦ Has a sub-table of calls, like bind, connect, etc.

## Plumbing

- ♦ Each message is put in a sk\_buff structure
- ♦ Between socket/application and device, the sk\_buff is passed through a stack of protocol handlers
  - † These handlers update internal bookkeeping, wrap payload in their headers, etc.
- At the bottom is the device itself, which sends/receives the packets

### sk\_buff

(from Understanding Linux Networking Internals)

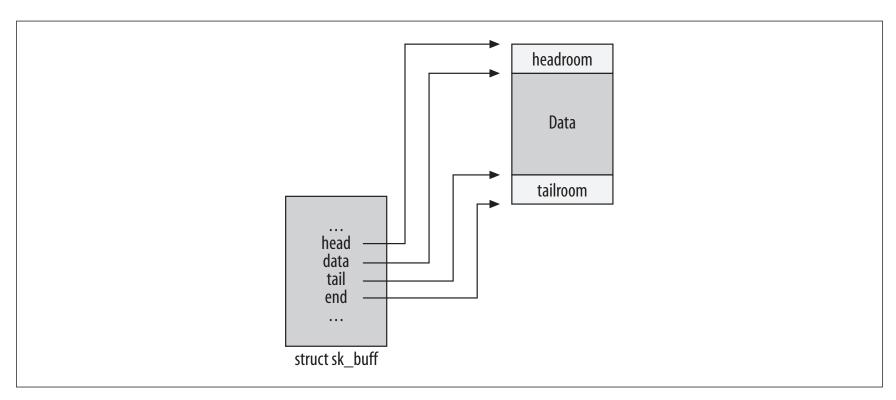


Figure 2-2. head/end versus data/tail pointers

## Efficient packet processing

♦ Moving pointers is more efficient than removing headers

♦ Appending headers is more efficient than re-copy



## Networking Part Two

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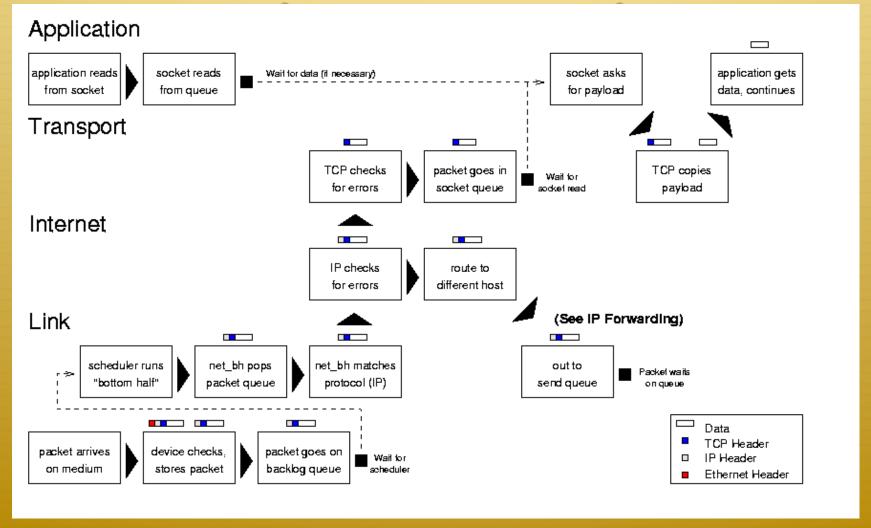
## Recap from last class

→ Layering

♦ L2, L3, L4 basics

♦ Packet walkthrough

## Walk through how a revd packet is processed



Source = http://www.cs.unh.edu/cnrg/people/gherrin/linux-net.html#tth\_sEc6.2

## Interrupt handler

- → "Top half" responsible to:
  - Allocate a buffer (sk\_buff)
  - ♦ Copy received data into the buffer
  - ♦ Initialize a few fields
  - ♦ Call "bottom half" handler
- ♦ In some cases, sk\_buff can be pre-allocated, and network card can copy data in (DMA) before firing the interrupt
  - ♦ Lab 6 will follow this design

## Quick review

- ♦ Why top and bottom halves?
  - → To minimize time in an interrupt handler with other interrupts disabled
  - ♦ Gives kernel more scheduling flexibility
  - \* Simplifies service routines (defer complicated operations to a more general processing context)

## Digression: Softirqs

- ♦ A hardware IRQ is the hardware interrupt line
  - Also used for hardware "top half"
- ♦ Soft IRQ is the associated software "interrupt" handler
  - ♦ Or, "bottom half"
- ♦ How are these implemented in Linux?
  - → Two canonical ways: Softirq and Tasklet
  - ♦ More general than just networking

## Softirqs

- ♦ Kernel's view: per-CPU work lists
  - ♦ Tuples of <function, data>
- ♦ At the right time, call function(data)
  - Right time: Return from exceptions/interrupts/sys. calls
  - Also, each CPU has a kernel thread ksoftirqd\_CPU# that processes pending requests
  - \* ksoftirqd is nice +19. What does that mean?
    - ♦ Lowest priority only called when nothing else to do

## Softirqs, cont.

- ♦ Device programmer's view:
  - ♦ Only one instance of a softirq function will run on a CPU at a time
    - ♦ Doesn't need to be reentrant
      - \* reentrant if it can be interrupted in the middle of its execution and then safely called again ("re-entered") before its previous invocations complete execution
    - ♦ If interrupted, won't be called again by interrupt handler
      - ♦ Subsequent calls enqueued!
  - ♦ One instance can run on each CPU concurrently, though
    - ♦ Must use locks

#### **Tasklets**

- ♦ For the faint of heart (and faint of locking prowess)
- ♦ Constrained to only run one at a time on any CPU
  - ♦ Useful for poorly synchronized device drivers
    - ♦ Say those that assume a single CPU in the 90's
  - → Downside: If your driver uses tasklets, and you have multiple devices of the same type---the bottom halves of different devices execute serially

# Softirq priorities

- \* Actually, there are 6 queues per CPU; processed in priority order:
  - # HI\_SOFTIRQ (high/first)
  - **♦** TIMER
  - ♦ NET TX
  - ♦ NET RX
  - **♦** SCSI
  - → TASKLET (low/last)

## Observation 1

- ♦ Devices can decide whether their bottom half is higher or lower priority than network traffic (HI or TASKLET)
  - \* Example: Video capture device may want to run its bottom half at HI, to ensure quality of service
  - ♦ Example: Printer may not care

### Observation 2

- ♦ Transmit traffic prioritized above receive. Why?
  - The ability to send packets may stem the tide of incoming packets
    - ♦ Obviously eliminates retransmit requests based on timeout
    - ♦ Can also send "back-off" messages

### Receive bottom half

- ♦ For each pending sk\_buff:
  - Pass a copy to any taps (sniffers)
  - ♦ Do any MAC-layer processing, like bridging
  - ♦ Pass a copy to the appropriate protocol handler (e.g., IP)
    - ♦ Recur on protocol handler until you get to a port
      - ♦ Perform some handling transparently (filtering, ACK, retry)
    - ♦ If good, deliver to associated socket
    - ♦ If bad, drop

## Socket delivery

- Once the bottom half/protocol handler moves a payload into a socket:
  - Check and see if the task is blocked on input for this socket
  - ♦ If so, wake it up
- ♦ Read/recv system calls copy data into application

# Socket sending

- ♦ Send/write system calls copy data into socket
  - ♦ Allocate sk\_buff for data
  - ♦ Be sure to leave plenty of head and tail room!
- System call does protocol handling during application's timeslice
  - ♦ Note that receive handling done during ksoftirqd timeslice
- ♦ Last protocol handler enqueues a softirq to transmit

#### Transmission

- ♦ Softirq can go ahead and invoke low-level driver to do a send
- ♦ Interrupt usually signals completion
  - ♦ Interrupt handler just frees the sk\_buff

# Switching gears

- ♦ We've seen the path network data takes through the kernel in some detail
- ♦ Now, let's talk about how network drivers handle heavy loads

## Our cup runneth over

- ♦ Suppose an interrupt fires every time a packet comes in
  - † This takes N ms to process the interrupt
- ♦ What happens when packets arrive at a frequency approaching or exceeding N?
  - ♦ You spend all of your time handling interrupts!
- ♦ Will the bottom halves for any of these packets get executed?
  - ♦ No. They are lower-priority than new packets

### Receive livelock

- ♦ The condition that the system never makes progress because it spends all of its time starting to process new packets
- \* Real problem: Hard to prioritize other work over interrupts
- ♦ Principle: Better to process one packet to completion than to run just the top half on a million

# Receive livelock in practice

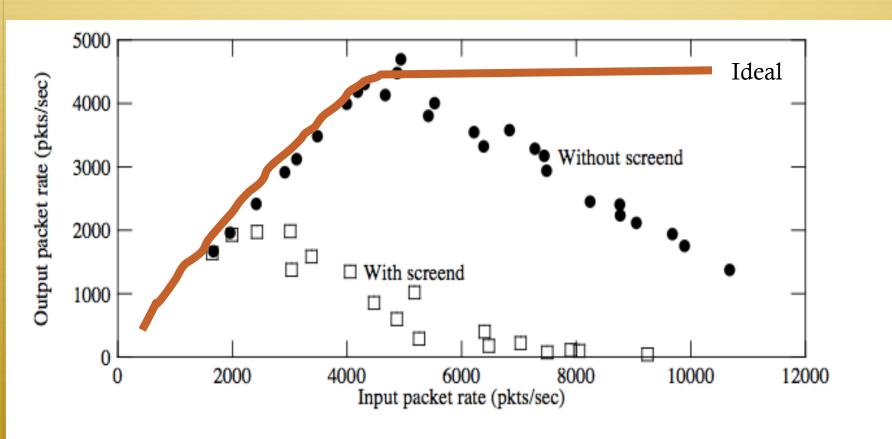


Fig. 2. Forwarding performance of unmodified kernel.

Source: Mogul & Ramakrishnan, ToCS 96

# Shedding load

- ♦ If you can't process all incoming packets, you must drop some
- Principle: If you are going to drop some packets, better do it early!
- ♦ If you quit taking packets off of the network card, the
  network card will drop packets once its buffers get full

## Idea

- ♦ Under heavy load, disable the network card's interrupts
- ♦ Use polling instead
  - \* Ask if there is more work once you've done the first batch
- ♦ This allows a packet to make it all the way through all of the bottom half processing, the application, and get a response back out
- ♦ Ensuring some progress! Yay!

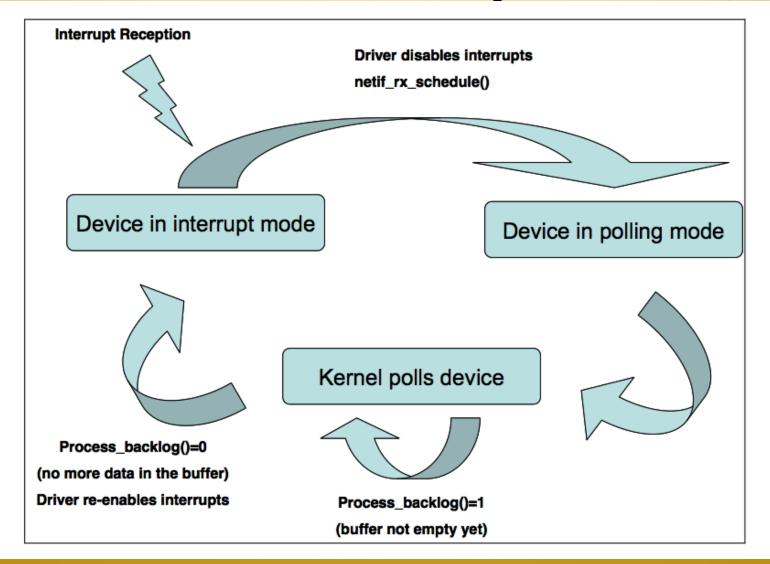
# Why not poll all the time?

- ♦ If polling is so great, why even bother with interrupts?
- ♦ Latency: When incoming traffic is rare, we want highpriority, latency-sensitive applications to get their data ASAP

# General insight

- ♦ If the expected input rate is low, interrupts are better
- ♦ When the expected input rate gets above a certain threshold, polling is better
- → Just need to figure out a way to dynamically switch between the two methods...

## Pictorially...



Source: download.intel.com/design/intarch/PAPERS/323704.pdf

# Why haven't we seen this before?

- ♦ Why don't disks have this problem?
- ♦ Inherently rate limited
- → If the CPU is bogged down processing previous disk requests, it can't issue more
- ♦ An external CPU can generate all sorts of network inputs

### Linux NAPI

- ♦ Or New API. Seriously.
- ♦ Every driver provides a poll() method that does the low-level receive
  - ♦ Called in first step of softirq RX function
- ♦ Top half just schedules poll() to do the receive as softirq
  - ♦ Can disable the interrupt under heavy loads; use timer interrupt to schedule a poll
  - ♦ Bonus: Some rare NICs have a timer; can fire an interrupt periodically, only if something to say!

#### NAPI

- ♦ Gives kernel control to throttle network input
- ♦ Slow adoption means some measure of driver rewriting
- ♦ Backwards compatibility solution:
  - ♦ Old top half still creates sk\_buffs and puts them in a queue
  - ♦ Queue assigned to a fake "backlog" device
  - ♦ Backlog poll device is scheduled by NAPI softirq
  - ♦ Interrupts can still be disabled

## NAPI Summary

- ♦ Too much input is a real problem
- ♦ NAPI lets kernel throttle interrupts until current packets processed
- ♦ Softirq priorities let some devices run their bottom halves before net TX/RX
  - ♦ Net TX handled before RX

## General summary

- ♦ Networking basics and APIs
- ♦ Idea of plumbing from socket to driver
  - ♦ Through protocol handlers and softirq poll methods
- ♦ NAPI and input throttling