Solving Capture in Switched Two-Node Ethernets by Changing Only One Node

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Background: History of Ethernet

- designed when 10 Mb/s was enough bandwidth to handle hundreds of hosts
- all hosts share one broadcast medium
 collisions can occur during transmission
 ⇒ need a way to resolve collisions
- binary exponential backoff resolves collisions between N hosts in time log N on average
- if N = 2, successive collisions can trick a host into backing off for too long, causing long delays and short-term unfairness

The Standard Ethernet Protocol

When there are packets to send, locally execute the following:

- 1. set *attempts* := 0. Remember this for later; the *attempts* counter is reset for *every* new packet.
- 2. wait for silence on the network...
- 3. Attempt transmission! If successful, wait a few bit-times and then go get another packet. Otherwise, a collision has occured...
- 4. increment attempts by 1
- 5. choose uniform random integer delay between 0 and $2^{attempts} 1$ inclusive.
- 6. Sleep for delay slot-times, where a slot is the time taken to transmit 512 bits.
- 7. Proceed to step 2.

Why Ethernet breaks with two busy hosts: The "Capture" Effect.

- Say Alice and Bob each have lots of packets.
- After k collisions, Alice wins and sends a packet while Bob goes back to sleep. Alice continues to send packets.
- Bob wakes up while Alice is still sending packets. They collide after Bob sees the end of Alice's pth packet.
- Alice chooses a delay between 0 and 1.
 Bob chooses a delay between 0 and 2^k-1.
 Guess who wins most of the time?

The Capture effect ... con't

- Alice wins, Bob goes to k + 1 collisions, and goes to sleep again
- Bob's odds of winning decrease exponentially with each collision
- When Alice runs out of packets, Bob's *attempts* counter is high and he's sleeping for a long time.
 - so the network is completely idle until he wakes up and starts sending.
 - If Alice gets any new packets, they will get sent even before Bob gets a chance to send his first packet.

Net effects: (no pun intended, of course)

- large "run lengths" resulting in large variance of delay (diagram)
- lots of time wasted idling after a run

This is BAAAAAD

- real-time stuff likes short, predictable delay. (*eg.* realtime audio, video, distributed computing, even just remote typing.)
- long delays confuse higher level protocols (and users!) into thinking something is wrong, when nothing is.

Some really basic ideas:

- Modify one node, the other runs standard Ethernet
- The standard Ethernet node normally hogs the network as long as it wants
- The modified node stops this using the electronic equivalent of a baseball bat
- When the modified node is transmitting, it can keep track of how many times the standard node has collided, *ie.* keep track of the standard node's collision counter.
- When Modified node's turn is over, let the standard node transmit unhindered for awhile.

SHEP

Switched Half-duplex Ethernet Protocol (slightly simplified)

 $\mathbf{S} = \mathbf{S}$ tandard Ethernet Node

The modified node runs the following:

- 0. note the wall-clock time at the beginning of my turn.
- 1. When there is a packet to transmit, wait-for-silence plus the interframe gap, then attempt transmission as in regular Ethernet.
- 2. If a collision occurs, always choose a backoff delay of 0...
- 3. ... until **S**'s collision counter reaches a fixed maximum maxCC, after an interval of time T.
- 4. The end of my turn has arrived. Concede control to \mathbf{S} . Be absolutely silent and let \mathbf{S} transmit unhindered for an interval of time T.
- 5. Proceed to step 0.

Note turn length is stochastic, maxCC is fixed.

- maxCC = 1 seems to be the best choice, because letting it get bigger increases delays and increases idle time between switch overs. Exception: long networks – a bit bigger, maybe 2 or 3.
- When to concede? Some choices:
 - 1. Immediately when **S**'s collision counter reaches maxCC
 - 2. Retry once more.
 - * use the idle time that's likely to occur immediately after this collision.
 - * strictly bounded delay
 - 3. "Transmit Damnit!" ⇐=
 - * "almost" bounded delay, slightly higher capacity.

Comparison to CABEB (& others)

- We change only one node, CABEB (BLAM, full-duplex Ethernet) both nodes need to be changed to get good results.
- Overload with CABEB against standard node ⇒ CABEB node hogs bandwidth, even if standard node offers higher load.
 [Tables 6.1-6.3 of Ramakrishnan & Yang]
- Since CABEB is one-packet-per-turn, packet sizes differ $s1, s2 \Rightarrow$ relative load forced to $\frac{s1}{s2}$ during periods of overload.
- However, SHEP has slightly less capacity for small packets, since extra collisions & idle time between turns.

Summary, Conclusions

With two busy nodes running standard Ethernet protocol, one node can "capture" the network, causing

- significant short-term unfairness
- frequent large delays

SHEP

- efficient, round-robin service
- decreases std. dev. of delay by 2 orders of magnitude
- eliminates large delays
- negligible cost in bandwidth