Selection criteria for the next generation of TCP

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Time horizon

TCP Reno (and friends) lasted 20 years
Design next generation to last as long

Scalability vital; reflect this in benchmarks

 1Gbps common, 1Tbps cutting edge

 Future backward compatibility

 Router-assisted has to be compatible with it

 Fix more problems than bandwidth-delay

 Wireless? Slow start?

TCP friendliness is medium term

- New protocol *must* work "OK" with Reno
 - E.g., neither should starve the other
- Time between OS upgrades is years not decades
 - Fairness with *itself* is more critical (e.g. RTT fairness)
 - RTT unfairness means even TCP is not TCP friendly

Benchmarking:

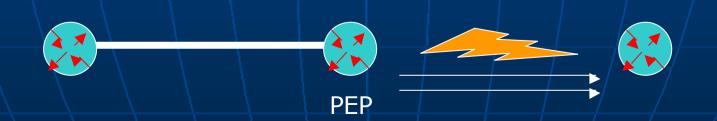
- background traffic Reno
- background traffic of the new protocol
- mixtures

and and

Wireless / lossy links

TCP Reno suffers on wireless links

- TCP solutions like Westwood, Veno, ...
- Workarounds like "performance enhancing proxies"
- Break end-to-end semantics (like encryption)
- Wireless Gigabit is already possible
 - Will have lossy links on fast long distance routes



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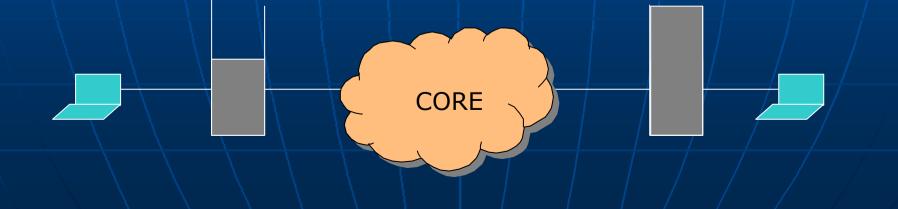
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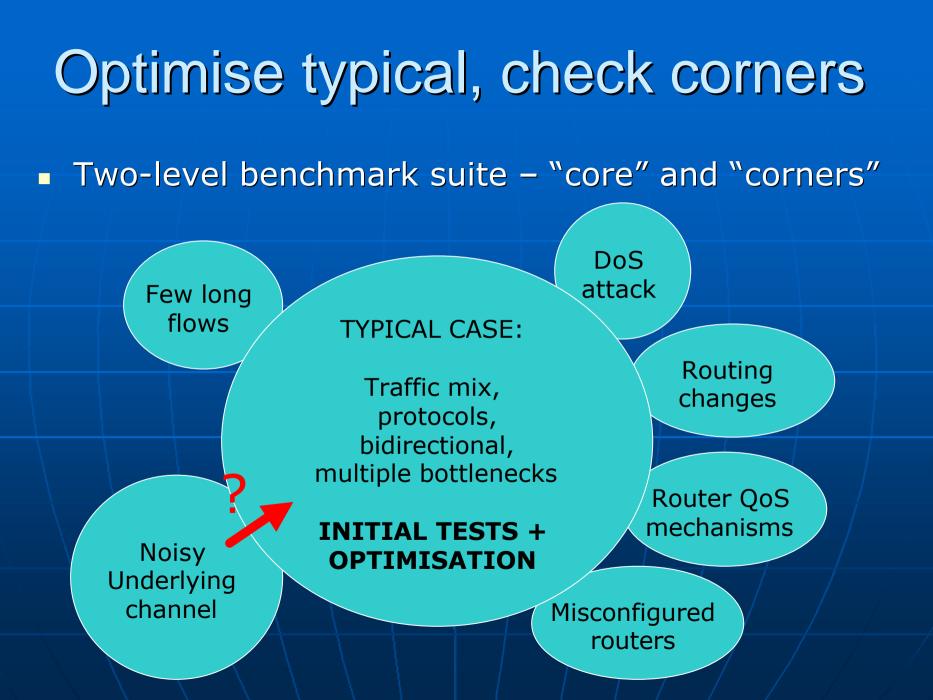
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 - Loss passed on to TCP
 - *Random delay* due to link layer retransmissions
- Wireless will use "multi-user diversity"
 - Wants packets from all flows buffered
 - Wired links just want packets from at least one flow
 - Further increase in jitter

Multiple bottlenecks

Rarely have a single bottleneck

- Typically access network at each end of the path
- Single bottleneck tests
 - illusion that buffer overflow = maximum observed RTT
 - over-emphasise synchronised loss
- Need balance
 - Single-link tests take less infrastructure, simulation time





WAN-in-Lab testbed

Dummynet and simulation introduce artifacts

Also need to test on *real* equipment

- WAN with real delays, located in a single room
 - Connected to an external WAN (Ultralight)
- Open for the community to use for benchmarking

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WAN-in-Lab capabilities

Current	Planned
Two 2.5G bottlenecks Multiple 1G bottlenecks	Six 2.5G bottlenecks
Two "real" delays (Emulate cross traffic delay)	Up to six "real" delays
End-to-end RTT, drop	Per-router delay, drop (movable DAG cards)

Conclusions

The new "TCP friendly" requirement
Response function should scale beyond now
Consider all of TCP's weaknesses
Not just fast long-distance networks
Multiple bottleneck topologies

Simulation, emulation and real networks
 WAN-in-Lab