Requirements for a CCID for Interactive Media no draft (yet)

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http://www.phelan-4.com/dccp/IETF-64-media-CCID.ppt

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DCCP IETF-64 CCID for Interactive Media

History

- Group has interest in interactive media apps
 - draft-burness-dccp-interactive-apps
 - draft-phelan-mfrc
 - draft-phelan-dccp-media (evolved to draft-ietf-dccp-tfrc-media)
- Some overlap with iccrg and tmrg
 - But in Paris we decided dccp was good place to center work
 - Start with requirements
- Mailing list discussions perhaps as precursor to draft

Some Possible Requirements

but not all, and other issues

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Delay

- Human-to-human voice apps can tolerate no more than about 150ms delay from lips to ears
 - The limit **is** one-way, not round trip
 - One-way 250ms, return 10ms doesn't work
- Human-to-human video limited by voice
 - Channel surfing limit around 500ms round trip
- But how do you phrase a requirement for the CCID?
 - Must not unnecessarily delay transmissions
 - Wishy-washy

TCP-Friendliness

- Competing flows with similar circumstances get order of magnitude equal throughput measured over time periods lasting seconds
- I think this is a requirement
 - DCCP is general-purpose transport, deployable anywhere in Internet
- This has gotten most discussion on mailing list
 - Can't we trade off our self-limiting discipline against TCP's grabwhat-you-can greed for a bigger than fair share?
 - It hurts us inelastic apps more than the elastic apps, so we should get more
 - How about measuring fairness over longer time periods?

TCP Courage

- Apps that self-limit to less than fair share shouldn't be driven off
 - Note that TFRC has this characteristic (at least for large packet flows)

Slow Start

- New flows must gently enter network
- Like TCP, TFRC slow start
 - Although not necessarily those exact algorithms

Variable Rate

- Combining Stop/Start and variable rate in this slide
 - Silence suppression for voice
 - Motion compensation for video
- Basic premise CCID should not force apps to use more bits now in order to protect capability to use more bits later
- Toughest problem, IMHO

What's Fair?

• How do you judge fair share?

– Peak rate must be <= fair share?</p>

- Average rate must be <= fair share?</p>
- TCP-Fairness judged on scale of seconds

– So average rate seems reasonable

- But peak rate must be limited too
 - Peak rate must be less than lowest link capacity

What's Fair? (2)

- Requirements (IMHO):
 - Peak rate <= fair share allowed</p>
 - Average rate <= fair share better</p>
 - Average over seconds
 - But peak rate must be limited also
 - Peak-to-average ratio may be limited
 - E.g., if actual average less than peak/X, use peak/X <= fair share

Small Packets

- Fairness judged in bytes/second, not packets/sec
- Bytes/sec are app data plus necessary DCCP and IP headers
 - Could also include MAC header
 - Small packets have more headers, so less app throughput

Return to High Rate

- In addition to reducing the allowed rate during congestion, the CCID must allow increasing the rate when congestion dissipates
- Apps may chose not to return to high rate
 - User perception of variable quality as worse than constant low quality