The Some Congestion Experienced ECN Codepoint

Abstract

   This memo reclassifies ECT(1) to be an early notification of
   congestion on ECT(0) marked packets, which can be used by AQM
   algorithms and transports as an earlier signal of congestion than
   CE. It is a simple, transparent, and backward compatible upgrade to
   existing IETF-approved AQMs, RFC3168, and nearly all congestion
   control algorithms.

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1. Terminology

   The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT",
   "SHOULD", "SHOULD NOT", "RECOMMENDED", "NOT RECOMMENDED", "MAY", and
   "OPTIONAL" in this document are to be interpreted as described in
   [RFC2119] and [RFC8174] when, and only when, they appear in all
   capitals, as shown here.

2. Introduction

Traditional TCP congestion control exhibits a "sawtooth" pattern which, in the most favourable cases, oscillates around the optimum operating point of maximum throughput and minimum delay, which exists at the point where the congestion window equals path BDP. The term "sawtooth" brings to mind the straight-edged graphs of TCP Reno, but the equally common TCP CUBIC is essentially similar in character, as are other AIMD-derived algorithms.

A number of proposals have sought to improve this, but introduce various other tradoffs in return. TCP Vegas is consistently outcompeted by standard TCPs, DCTCP proved to be too aggressive for deployment in the public Internet, and while BBR appears to have avoided both of these problems, its complexity makes it difficult to implement correctly. Each of these proposals is characterised by primarily changing only the endpoints, not the network nodes on the path between them; though DCTCP is intended for use with a specific style of AQM, it can work with standard AQMs as long as there is no competing non-DCTCP traffic.

Some other proposals have attempted to convey information about the network path explicitly, by having network nodes inject data about link capacity and/or utilisation into passing traffic. These proposals have generally been unsuccessful due to the complex slow-path processing required in network nodes, and are not widely deployed. The only successful proposal of this type is Explicit Congestion Notification [RFC3168] which allows an AQM to signal congestion by marking packets with (essentially) a one-bit signal in preference to dropping them.

ECN defines a two-bit field supporting four codepoints, of which three are in active use and the fourth is a semantic duplicate. It was explicitly suggested during ECN's development that new meaning could be given to this spare codepoint, including as a lesser indication of congestion. With an alternative use of this codepoint having fallen out of favour, the time is right to revisit this suggestion and propose a workable method of applying it.

In so doing, care must be taken that backwards compatibility is maintained with existing traffic, endpoints and network nodes that are known or suspected to have been deployed. Keeping the changes to on-wire protocols minimal, and the complexity of implementation low, are also highly desirable.

This memo reclassifies ECT(1) to be an early notification of congestion on ECT(0) marked packets, which can be used by AQM algorithms and transports as an earlier signal of congestion than CE ("Congestion Experienced").

This memo also briefly discusses how transports should respond to
ECT(1) marked packets. Detailed specifications of this behaviour are
left to transport-specific memos.

3. Background

[RFC3168] defines the lower two bits of the (former) TOS byte in the
IPv4/6 header as the ECN field. This may take four values: Not-ECT,
ECT(0), ECT(1) or CE.

Binary Keyword References

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 00     Not-ECT (Not ECN-Capable Transport)     [RFC3168]
 01     ECT(1) (ECN-Capable Transport(1))       [RFC3168]
 10     ECT(0) (ECN-Capable Transport(0))       [RFC3168]
 11     CE (Congestion Experienced)             [RFC3168]
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Research has shown that the ECT(1) codepoint goes essentially
unused, with the "Nonce Sum" extension to ECN having not been
implemented in practice and thus subsequently obsoleted by [RFC8311]
(section 3). Additionally, known [RFC3168] compliant senders do not
emit ECT(1), and compliant middleboxes do not alter the field to
ECT(1), while compliant receivers all interpret ECT(1) identically
to ECT(0). These are useful properties which represent an
opportunity for improvement.

Experience gained with 7 years of [RFC8290] deployment in the field
suggests that it remains difficult to maintain the desired 100% link
utilisation, whilst simultaneously strictly minimising induced delay
due to excess queue depth - irrespective of whether ECN is in use.
This leads to a reluctance amongst hardware vendors to implement the
most effective AQM schemes because their headline benchmarks are
throughput-based.

The underlying cause is the very sharp "multiplicative decrease"
reaction required of transport protocols to congestion signalling
(whether that be packet loss or CE marks), which tends to leave the
congestion window significantly smaller than the ideal BDP when
triggered at only slightly above the ideal value. The availability
of this sharp response is required to assure network stability (AIMD
principle), but there is presently no standardised and backwards-
compatible means of providing a less drastic signal.

4. Some Congestion Experienced

As consensus has arisen that some form of ECN signaling should be an
earlier signal than drop, this memo changes the meaning of ECT(1) to
SCE, meaning "Some Congestion Experienced". Since there is no longer

ambiguity between two ECT codepoints, ECT(0) is referred to as ECT.
The ECN-field codepoint table then becomes:

Binary Keyword References

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00   Not-ECT (Not ECN-Capable Transport)   [RFC3168]
01   SCE (Some Congestion Experienced)   [This Internet-draft]
10   ECT (ECN-Capable Transport)   [RFC3168]
11   CE (Congestion Experienced)   [RFC3168]

This permits middleboxes implementing AQM to signal incipient
congestion, below the threshold required to justify setting CE, by
converting some proportion of ECT codepoints to SCE ("SCE marking").
Existing [RFC3168] compliant receivers MUST transparently ignore
this new signal with respect to congestion control, and both
existing and SCE-aware middleboxes SHOULD convert SCE to CE in the
same circumstances as for ECT, thus ensuring backwards compatibility
with [RFC3168] ECN endpoints.

Permitted ECN codepoint packet transitions by middleboxes are:

     Not-ECT ->   Not-ECT (or drop)
     ECT   ->   ECT or SCE or CE
     SCE   ->   SCE or CE
     CE   ->   CE

In other words, for ECN-aware flows, the ECN marking of an
individual packet MAY be increased by a middlebox to signal
congestion, but MUST NOT be decreased, and packets SHALL NOT be
altered to appear to be ECN-aware if they were not originally, nor
vice versa. Note however that SCE is numerically less than ECT, but
semantically greater, and the latter definition applies for this
rule.

Receivers and transport protocols conforming to this specification
SHALL continue to apply the [RFC3168] interpretation of the CE
codepoint, that is, to signal the sender to back off send rate to
the same extent as if a packet loss were detected. This maintains
compatibility with existing middleboxes, senders and receivers.

New SCE-aware receivers and transport protocols SHOULD interpret the
SCE codepoint as an indication of mild congestion, and respond
accordingly by applying send rates intermediate between those
resulting from a continuous sequence of ECT codepoints, and those
resulting from a CE codepoint. The ratio of ECT and SCE codepoints
received indicates the relative severity of such congestion, with a
higher proportion of SCE codepoints indicating more congestion.

The intent of SCE marking is a "cruise control" signal which permits middleboxes to request relatively small reductions in send rate, or merely a slowing of send rate growth. Accordingly, SCE marks SHOULD progressively trigger exit from exponential slow-start growth, then reduction to Reno-linear growth (for congestion control algorithms which support higher growth rates in congestion-avoidance phase), then a halt to send rate growth, then a gradual reduction of send rate. For immediate large reductions of send rate, the CE mark MUST retain its original Multiplicative Decrease power as per [RFC8511], and compliant AQMs SHOULD retain the ability to employ it where appropriate.

Details of how to implement SCE awareness at the transport layer will be left to additional Internet Drafts yet to be submitted. To ensure RTT-fair convergence with single-queue SCE AQMs, transports SHOULD stabilise at lower SCE-mark ratios for higher BDPs, and MAY reduce their response to CE marks IFF they are responding to SCE signals received at around the same time (eg. within 1-2 RTTs) in the same flow.

To maximise the benefit of SCE, middleboxes SHOULD begin to produce SCE marks at lower congestion levels than they begin to produce CE marks. This will usually ensure that SCE-aware flows avoid receiving CE marks. When a single-queue AQM is upgraded to SCE awareness, this will tend to cause SCE flows to give way to non-SCE flows; to avoid this behaviour, single-queue AQMs MAY be left as RFC-3168 compliant without SCE support.

For the avoidance of doubt, a decision to mark CE or to drop a packet always takes precedence over SCE marking.

5. Examples of use

5.1. Codel-type AQMs

A simple and natural way to implement SCE in a Codel-type AQM is to mark all ECT packets as SCE if they are over half the Codel target sojourn time, and not marked CE by Codel itself. This threshold function does not necessarily produce the best performance, but is very easy to implement and provides useful information to SCE-aware flows, often sufficient to avoid receiving CE marks whilst still efficiently using available capacity.

For a more sophisticated approach avoiding even small-scale oscillation, a stochastic ramp function may be implemented with 100% marking at the Codel target, falling to 0% marking at or above zero sojourn time. The lower point of the ramp should be chosen so that SCE is not accidentally signalled due to CPU scheduling latencies or serialisation delays of single packets. Absent rigorous analysis of

these factors, setting the lower limit at half the Codel target
should be safe in many cases.

The default configuration of Codel is 100ms interval, 5ms target. A
typical ramp function for these parameters might cease marking below
2.5ms sojourn time, increase marking probability linearly to 100% at
5ms, and mark at 100% for sojourn times above 5ms (in which CE
marking is also possible).

In single-queue AQMs, the above strategy will result in SCE flows
yielding to pressure from non-SCE flows, since CE marks do not occur
until SCE marking has reached 100%. A balance between smooth SCE
behaviour and fairness versus non-SCE traffic can be found by having
the marking ramp cross the Codel target at some lower SCE marking
rate, perhaps even 0%. A two-part ramp, reaching 1/sqrt(X) at the
Codel target (for some chosen X, a cwnd at which the crossover
between smoothness and fairness occurs) and ramping up more steeply
thereafter, has been implemented successfully for experimentation.

The CNQ algorithm [I-D.morton-tsvwg-cheap-nasty-queueing] offers a
relatively simple way to limit this yielding behaviour and ensure
that, even in competition with non-SCE flows, SCE flows maintain a
reasonable minimum throughput capability. This may be sufficient to
avoid the need for the two-part ramp described above.

Flow-isolating AQMs, including especially CNQ and DRR++ based
algorithms, should avoid signalling SCE to flows classified as
"sparse", in order to encourage the fastest possible convergence to
the fair share.

5.2. RED-type AQMs (including PIE)

There are several reasonable methods of producing SCE signals in a
RED-type AQM.

The simplest would be a threshold function, giving a hard boundary
in queue depth between 0% and 100% SCE marking. This could be a
sensible option for limited hardware implementations. The threshold
should be set below the point at which a growing queue might trigger
CE marking or packet drops.

Another option would be to implement a second marking probability
function, occupying a queue-depth space just below that occupied by
the main marking probability function. This should be arranged so
that high marking rates (ideally 100%) are achieved at or before the
point at which CE marking or packet drops begin.

For PIE specifically, a second marking probability function could be
added with the same parameters as the main marking probability
function, except for a lower QDELAY_REF value. This would result in

the SCE marking probability remaining strictly higher than the CE
marking probability for ECT flows.

5.3. TCP

Some mechanism should be defined to feed back SCE signals to the
sender explicitly. Details of this are left to [I-D.grimes-tcpm-
tcpsce]; use could be made of the redundant NS bit in the TCP
header, which was formerly associated with ECT(1) in the Nonce Sum
specification.

The recommended response to each single segment marked with SCE is
to reduce cwnd by an amortised 1/sqrt(cwnd) segments. Other
responses, such as the 1/cwnd from DCTCP, are also acceptable but
may perform less well.

5.4. Other

New transports under development, such as QUIC, may implement a
fine-grained signal back to the sender based on SCE. QUIC itself
appears to have this sort of feedback already (counting ECT(0),
ECT(1) and CE packets received), and the data should be made
available for congestion control.

6. Compatibility

6.1. Existing ECN & AQM Deployments

SCE explicitly retains [RFC8511] compliant Multiplicative Decrease
responses to CE marks, and conventional Multiplicative Decrease
responses to packet loss. SCE senders' behaviour is thus naturally
compliant with existing specifications when running over existing
networks.

Existing endpoints, supporting Not-ECT or [RFC3168] compliant
congestion control, are required to treat SCE marks (that is,
ECT(1)) as identical to ECT(0), and will thus transparently ignore
SCE marks. This is allowed for in SCE's design, and allows SCE
middleboxes to be deployed into a heterogeneous network.

Hence the incremental deployability of SCE endpoints and middleboxes
is good.

6.2. L4S

L4S also claims the ECT(1) codepoint, with significantly different
semantic meaning than SCE. In the L4S system, ECT(1) is used to
identify L4S flows, to distinguish them from [RFC3168] flows -
necessary since in L4S, the semantic meaning of CE marks is also
changed.

Since L4S connections are explicitly negotiated through support of AccECN, and AccECN doesn't support SCE, there is no ambiguity regarding the mode of the connection as far as endpoints are concerned.

SCE middleboxes will treat L4S flows in the same way as [RFC3168] does.

L4S middleboxes may interpret ECT packets which have received SCE markings at some other SCE-aware middlebox as though they were L4S traffic. This may result in a higher CE marking rate and/or different queuing behaviour. Though undesirable, this appears to be safe from SCE's point of view. Since the steady-state rate of SCE marking is likely to be low, the impact on L4S is also likely to be tolerable.

Accordingly, it appears as though the two experiments can coexist.

However, there is a secondary concern brought about by the L4S use of ECT(1) as a traffic identifier. If, as presently seems likely, it is found necessary to firewall L4S traffic off from the general Internet, then SCE-marked packets are also likely to be dropped at this boundary. This could have a significantly detrimental effect on ECT traffic traversing both an SCE and an L4S enabled network, even if the endpoints are not explicitly SCE aware.

7. Related Work

[RFC8087][RFC7567][RFC7928][RFC8290][RFC8289][RFC8033][RFC8034]

8. IANA Considerations

There are no IANA considerations.

9. Security Considerations

An adversary could inappropriately set SCE marks at middleboxes he controls to slow down SCE-aware flows, eventually reaching a minimum congestion window. However, the same threat already exists with respect to inappropriately setting CE marks on normal ECN flows, and this would have a greater impact per mark. Therefore no new threat is exposed by SCE in practice.

An adversary could also simply ignore SCE marks at the receiver, or ignore SCE information fed back from the receiver to the sender, in an attempt to gain some advantage in throughput. Again, the same could be said about ignoring CE marks, so no truly new threat is exposed. Additionally, correctly implemented SCE detection may actually improve long-term goodput compared to ignoring SCE.

An adversary could erase congestion information by converting SCE marks to ECT or Not-ECT codepoints, thus hiding it from the receiver. This has equivalent effects to ignoring SCE signals at the receiver. An identical threat already exists for erasing congestion information from CE marked packets, and may be mitigated by AQMs switching to dropping packets from flows observed to be non-responsive to CE.

An adversary could drop SCE-marked packets, believing them to be bogons (see also L4S Compatibility, above). Endpoints should be able to recover from this through retransmission and a reduction of cwnd. However, it is possible for this to lead to a significant denial of service. A workaround is to disable ECN for connections over the affected path.

10. Acknowledgements

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