Improving DCTCP/Prague Responsiveness

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Abstract

This paper shows that Data Center TCP (DCTCP) introduces unnecessary lag equivalent to 1.5–2 RTTs, due to the way it processes congestion feedback. A per-ACK moving average is proposed that cuts out 1 RTT of this lag. Paradoxically, the rest of the lag is reduced by spreading the congestion response over a round. The EWMA still inherently smooths over the same set number of round trips, even though it is clocked per-ACK.

1 Problem

Common implementations of DCTCP [AGM⁺10] (e.g. in Windows, Linux, or FreeBSD [BTB⁺17]) take two rounds before a change in congestion on the path fully feeds into the moving average that regulates its response. This is on top of the inherent round trip of delay in the feedback loop.

Both extra rounds are due to the two-stage process for maintaining the EWMA then using it to respond to congestion (see Figure 1). The first stage introduces a round of delay (RTT_i) while it accumulates the marking fraction before it can calculate the EWMA (α_i) , which it passes to the second stage that uses the EWMA to reduce the congestion window (W) by $\alpha_i W_i$. The second extra round of delay is because the first stage clocks independently, while the second stage is triggered by the arrival of congestion feedback (a red-coloured ACK).

So it takes two rounds before a full round of the congestion that triggered the start of the second stage has fed through into the EWMA that the first stage passes to the second. This is further exacerbated by entering congestion window reduced (CWR) state as soon as the initial inadequate response has been applied, which suppresses any further response for a round, while all the congestion feedback arrives. Also, the lagged congestion response will tend to overrun into the subsequent round, causing undershoot.

The problem seems to boil down to how to update the EWMA of marking on a per-ACK basis. But, more specifically, the problem is how to ensure that the time constant over which this per-ACK EWMA smooths itself keeps to a set number of rounds, even though the number of packets per round varies.

2 Per-ACK EWMA

Instead of the EWMA of the marking probability being upscaled by a constant factor, it is proposed to upscale it by flight_/ gain, where flight_ is the packets in flight, and gain is a constant (gain < 1). Then, as shown below, the EWMA can be maintained by a single continuous set of repetitive increments or decrements determined on each ACK.

As we shall see, scaling up the EWMA by flight_enables the repetitive per-packet operations to *implicitly* smooth the EWMA with a characteristic timescale proportionate to the RTT (specifically, RTT/gain).

This contrasts with the two-stage process of accumulating the marking fraction for a round before feeding it into an EWMA that is *explicitly* clocked per round.

The scaled up EWMA is effectively a smoothed count of the number of congestion marks per RTT, but scaled up by the constant 1/gain. The number of marks per RTT (v) is related to the marking probability (p) by v = p*flight.

Classical congestion controls suppress any further response for a round because their initial response is large and fixed. It seems wrong for DCTCP to mimic the timing of a classical congestion response, when it does not mimic its size. On first onset of congestion, DCTCP immediately responds with a tiny reduction (based on the previous absence of congestion), but then perversely it suppress any further response for a round trip.

So, once we have an EWMA of congestion marks that is updated continually on every ACK, it becomes possible to spread the reduction over the round. Then, if congestion continues to rise during the round, the EWMA will grow, and the response can pick up this growth as it proceeds.

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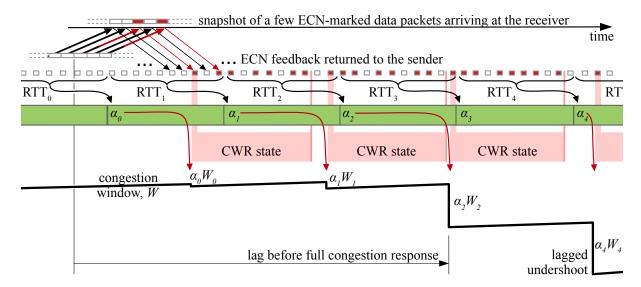


Figure 1: The problem: DCTCP's two stages for processing congestion feedback: 1) gathering feedback in a fixed sequence of rounds (RTT_i) to calculate the EWMA (α_i); 2) applying this EWMA on the first feedback mark, when it has had no time to gather enough feedback, which leads to a typically inadequate congestion response before entering congestion window reduced (CWR) state, which suppresses any further response for a round. See text for full commentary.

Definitions of variables

g = 1/gain. By default in DCTCP g = 16;
av_up : EWMA of the marks per round upscaled
 by g; Alternatively, it might help to think of
 this as the EWMA of the marking probability
 (alpha in DCTCP) upscaled by g * flight_.
flight_: the number of packets in flight when the
 marking probablity was fed into the EWMA
 (used for explanation, but not in the code);
flight : the number of packets in flight now;
ce_fb = 1 if ECN feedback per pkt; 0 otherwise.

2.1 Intuition

In DCTCP, the EWMA, alpha, is maintained per round trip as follows (in floating point arithmetic):

$$alpha += (F - alpha)/g,$$

where F is the fraction of marked bytes accumulated over the last round trip.

This can be approximated (see Appendix A) by repeatedly updating the EWMA on the feedback of every packet, but scaling down each update by the number of packets in that round, flight. That is, the following per-packet update:

The above per-packet update of the EWMA is roughly equivalent to the following per-packet update of the upscaled EWMA, av_up:

2.2 Implementation

2.2.1 Maintaining the EWMA

The ce_fb term can be implemented by adding 1 to av_up on feedback of each CE-marked packet. Given av_up is upscaled by g, this is equivalent to adding 1/g of a mark.

The number of packets acknowledged in the current round is flight. So repeatedly subtracting av_up/(flight_*g) on the arrival of every ACK would reduce av_up by av_up*flight/(flight_*g) in a round (see Appendix A). This approximates to av_up/g per round.

This is achieved as shown in the C-like pseudocode above by the repetitive additions, comparisons and decrements of a variable called av_carry, which carries forward the remainder between ACKs.

We have defined the general-purpose function repetitive_div_1() for the repetitive division, which will be reused later for maintaining the congestion window as well.

Figure 2 compares toy simulations of the above EWMA and the DCTCP EWMA (without changing cwnd). It can be seen that, whenever marks arrive, the algorithm always moves immediately, whereas DCTCP's EWMA does nothing until the next round trip cycle.

Per-ACK EWMA Update

Compared with DCTCP alpha av_up/cwnd/g — dctcp_alpha 0.06 0.05 0.04 0.03 0.02 0.01 0 0 100 200 300 400 500 600 700

Figure 2: Initial verification of per-ACK EWMA algorithm (with constant cwnd).

Packet no. (cwnd fixed at 50)

2.2.2 Responding to Congestion

The approach proposed in this section is not necessarily the final word on how to use the per-ACK EWMA for scalable congestion control (see § 6 for further ideas). Nonetheless, as a first step, we build incrementally on DCTCP, using its teaching selectively, but not departing too far from its intent.

DCTCP reduces cwnd by alpha*cwnd/2 in any round trip in which CE feedback is present. The proposed approach reduces cwnd by half the average number of marked packets per round, or av_up/(2*g). This is broadly equivalent to DCTCP in that it maintains the scalable '1/p' response function, except the following two slight differences could potentially alter the steady-state outcome:

- The window reduction is taken as a proportion of what the window has been in recent rounds, not what it is now (as in DCTCP) (see § 3).
- The window reduction is taken as a proportion of the amount of the window that has been *used* in recent rounds, not the maximum that the flow was entitled to use, i.e. packets in

flight not cwnd. Thus, if an application-limited flow has only used a quarter of the available window in recent rounds, the proposed reduction of cwnd will be only a quarter of that applied by DCTCP (see § 3.2).

The main departure from DCTCP is in the speed of response. Rather than reduce cwnd on the first sign of CE feedback then suppress further response for a round trip, it is proposed to spread the reduction over the round following the first sign of CE feedback. In other words, use the whole round while in CWR state to reduce cwnd as the EWMA updates, such that, by the end of the round it will still have reduced as much as it would have done if the whole reduction had been applied at the end.

This will exploit the fact that the EWMA (av_up) is continually updated on every ACK. So, at one extreme, if the first CE mark is immediately followed by many others, the EWMA will rapidly increase early in the round of CWR, and cwnd can be rapidly decreased accordingly. While, at the other extreme, if the first CE mark is the only CE mark in the round, cwnd will still have reduced by alpha*cwnd/2 by the end of the round, but the EWMA will hardly have increased above the value it took when the CWR round started.

To spread the reduction over the round, the proposed algorithm below does not divide the round into an arbitrary number of points where cwnd is altered by varying amounts. Instead, it decrements cwnd as soon as the algorithm first calculates that at least one packet of movement is possible.

```
On_each_ACK'd_packet {
    av_up += ce_fb - repetitive_div_1(
              av_up, flight*g, &av_carry);
    if (!cwr && ce_fb) {
        // Record start of CWR state
        next_seq = snd_next;
        cwr = true;
    if (cwr) {
        // Check still in CWR round
        if (snd_una < next_seq) {</pre>
            // Multiplicative Decrease
            cwnd -= repetitive_div_1(
          av_up, flight*g*2, &cwnd_carry);
        } else {
            cwr = false;
    }
```

As in DCTCP, the EWMA is calculated continuously, but it is only used if there is actual conges-

tion marking, when it is applied for one round of CWR.

CWR state then takes on a meaning that is nearly the opposite of its classical meaning. It no longer means 'congestion window reduced; no further reduction for a round'. Instead it means 'congestion window reduction in progress during this round'.

Although the motivation for this algorithm was not to prevent the stall caused by sudden increase in cwnd, it would probably serve to address this problem as well. Therefore it should supplant proportional rate reduction (PRR [MD13]), at least when responding to ECN.

Details of cwnd processing are omitted from the pseudocode if they are peripheral to the proposed changes. For instance, in real code cwnd would be prevented from falling below a minimum (default 2 segments), and the slow-start threshold would track reductions in cwnd (but see § 6 for alternative ideas). Also, the code would have to handle acknowledgement of multiple packets at a time (potentially with different congestion markings). It might also handle ECN markings on packets of different sizes, but the Linux implementation of DCTCP works well enough without attending to this degree of detail.

Also, for clarity, various integer arithmetic tricks are omitted from the pseudocode, such as choosing a value of g that is a power of 2, so that multiplication by g can be implemented with a bit-shift.

The pseudocode does, nonetheless, inherently attend to details such as loss of precision due to integer truncation. And note that cwnd is not updated on the ACK that ends CWR state, because it is updated on the marked ACK that starts CWR state.

2.2.3 The Whole AIMD Algorithm

For completeness, the pseudocode below includes a Reno-like additive increase. This is intended for periods when the congestion control is close to its operating point (if it is not, see § 6).

Unlike DCTCP, the proposed algorithm does not suspend additive increase during CWR state, with the following reasoning. DCTCP-like congestion controls are designed to induce roughly 2 ECN marks per round trip in steady state, so RTTs without marks are not meant to happen. Confining increase to periods that are not meant to happen creates an internal conflict within DCTCP's own design. Then, the algorithm's only escape is to store up enough decrease rounds to make space for a compensating period of increase. This has been found to cause unnecessary queue variation.

Instead, we continue additive increase regardless of CWR state. In place of suspending additive increase for a whole round, a fractional increase is calculated per-ACK (as in most TCP implementations) but it is skipped if an ACK carries congestion feedback. This thins down the additive increase as congestion rises (see § 3.1 of [BDS17]).

Also unlike DCTCP, the proposed algorithm increases cwnd by one segment over the actual window of packets in flight, not over the congestion window (which might not be fully used).

```
On_each_ACK'd_packet {
    // Update EWMA
    av_up += ce_fb - repetitive_div(false,
            av_up, flight*g, &av_carry);
    if (!ce_fb) {
        // Additive Increase
        cwnd += repetitive_div(false,
          g*2, flight*g*2, &cwnd_carry);
    } else if (!cwr) {
        // Record start of CWR state
        next_seq = snd_next;
        cwr = true;
    if (cwr) {
        // Check still in CWR round
        if (snd_una < next_seq) {</pre>
            // Multiplicative Decrease
            cwnd -= repetitive_div(true,
          av_up, flight*g*2, &cwnd_carry);
        } else {
            cwr = false;
        }
    }
// Repeated execution outputs true with
    frequency proportional to num/denom
// If subtract==true, flip direction of carry
bool repetitive_div(bool subtract,
          u32 num, u32 denom, u32 *carry) {
    _outcome = false;
    if (subtract) *carry = denom - *carry;
    *carry += num;
    if (*carry >= denom) {
        *carry -= denom;
        _outcome = true;
    }
    if (subtract) *carry = denom - *carry;
    return _outcome;
```

The additive increase stores its remainder in the same *cwnd_carry variable as the multiplicative decrease. So both numerator and denominator are

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scaled up such that AI uses the same denominator parameter as MD, otherwise the upscaling of the carry variable would be different.

This also requires the repetitive division function to be modified, to be able to either add or subtract the carry. To avoid ambiguity, we have used a slightly different function name, repetitive_div().

The new repetitive_div() function has an additional first boolean parameter subtract. When false, the function behaves no differently to the original repetitive_div_1() function. When true, it flips the direction of the number space holding the *carry variable, then flips it back again before terminating. This is preferable to switching between addition and subtraction when altering the carry, which would halve the number space available for the scaled up variables.

3 (Non-)Concerns

3.1 Circular Dependency?

There seems to be a circular dependency, because av_up is both upscaled by flight_ then used to update cwnd, which determines flight.

In fact, av_up is upscaled by flight_ (note the trailing underscore), which is what flight was at the time of each repetitive decrement or increment of av_up. So av_up depends on an implicit exponentially weighted moving average of flight with a characteristic smoothing timescale of g round trips. This removes any circular dependency.

3.2 Advantage to Application Limited Flows?

The reduction to cwnd is spread over the flight packets in a round of CWR, so each reduction is scaled down by flight in the denominator of the call to repetitive_div(). This means that the decrease of cwnd is actually by a multiplicative factor of flight, not of cwnd.

If a flow is not application-limited, the two amount to the same thing. But for app-limited flows, flight can be lower than cwnd, so the reduction in cwnd will be lower.

If a flow is only using a fraction of its congestion window, but it is still experiencing congestion, there is an implication that other flow(s) must have filled the capacity that the app-limited flow is 'entitled' to but not using. Then, it could be argued that the other flows have a higher cwnd than they

are 'entitled' to, so that the app-limited flow can reduce its 'entitlement' (cwnd) less than these other flows in response to congestion.

If this argument is not convincing, the reduction in cwnd could be scaled up by cwnd/flight. However, it is believed that the code is reasonable, perhaps better, as it stands.

Similar arguments can be used to motivate additive increase over the actual number of packets in flight, rather than the potential congestion window.

4 Evaluation Plan

For research purposes, we ought not to introduce two changes at once, without evaluating each separately. Therefore, initially, we ought to use the continually updated EWMA, av_up to reduce cwnd in the classical way. That is, on the first feedback of a CE mark, reduce cwnd once by av_up/(2*g). Then suppress further response for a round (CWR state). This should remove one round of lag (originally spent accumulating the marking fraction), but not the rest (spent reducing cwnd in response to a single mark, then doing nothing for a round while the extent of marking is becoming apparent).

Initial experiments will need to compare how quickly a DCTCP flow in congestion avoidance can reduce in response to a newly arriving flow or a reduction in capacity.

Initially the same gain as DCTCP (1/16) ought to be used. But it is possible that the reason DCTCP's gain had to be so low was because of the two rounds of built in lag in the algorithm. Therefore, it will be interesting to see if the gain can be increased (from 1/16 to 1/8 or perhaps even 1/2 ought to be tried).

The thinking here is that a fixed amount of lag in a response is not the same as smoothing. Lag applies the same response by later. Smoothing spreads the response out, adding lag to the end of the response, but not to the start. Given every DCTCP flow's response to each change has been lagged by 2 rounds, it is possible that all flows have had to be smoothed more than necessary, in order to prevent the excessively lagged responses from causing over-reactions and oscillations.

Our motivation for improving DCTCP's responsiveness is to ensure established flows yield quickly when new flows are trying to enter the system, without having to build a queue. However, it is possible that improved responsiveness will help address the incast problem, which will not have been helped by up to two rounds of unnecessary lag. Therefore, incast experiments could be of interest.

It will also be necessary to check performance in the following cases that might expose poor approximations in the algorithm (relative to DCTCP):

- 1. When the packets in flight has been growing for some time;
- 2. ... or shrinking for some time;
- 3. When the flow is application limited, with packets in flight varying wildly, rather than tracking the smoother evolution of cwnd.

Bob: Outcome of the evaluations to be added here.

5 Related Work

Reducing cwnd in one RTT by half of the marks per round trip (av_up/2/g) is similar but not the same as Relentless TCP [Mat09], which reduces cwnd by half a segment on feedback of each CE-marked packet. The difference is that av_up is a moving average, so it does not depend on the number of marks in any specific round, whereas the Relentless approach does. Relentless was designed for the classical approach with smoothing in the network, so it immediately applies a full congestion response without smoothing. In contrast, using the moving average implements the smoothing in the sender.

Like DCTCP, the per-ACK congestion response proposed in section 5.2 of [AJP11] maintains an EWMA of congestion marking probability, alpha. But, unlike DCTCP, it reduces cwnd by half of alpha (in units of packets) on feedback of each ECN mark. This is partway between Relentless and DCTCP, because it uses the smoothed average of marking, but it applies it more often in rounds with more marks. This still causes considerable jumpiness, because marks tend to be bunched into one round then clear for a few rounds, particularly with step-marking. In contrast, the approach proposed in the present paper limits the reduction within any one round to the averaged number of marks per round (as DCTCP itself does).

6 Ideas for Future Work

An EWMA of a queue-dependent signal is analogous to an integral controller. It filters out rapid variations in the queue that do not persist, but it also delays any response to variations that do persist. Faster control of dynamics should be possible by adding a proportional element, to create

a proportional-integral (PI) controller within the sender's congestion control. The proportional element would augment any reduction to the congestion window dependent on the rate of increase in av_up.

Separately, it would be possible to use the per-ACK EWMA of marks per round (av_up) as a good indicator of whether a flow has lost its closed-loop control signal, for instance because another flow has left the bottleneck, or capacity has suddenly increased. A flow could then switch into a mode where it searches more widely for a new operating point, for instance using paced chirping [MB19, § 3]. To deem that the closed loop signal had significantly slowed, it might calculate the average distance between marks implied by the EWMA av_up, multiply this by a heuristic factor, then compare this with the number of packets since the last mark. Alternatively, it might detect when the EWMA of the marks per round had reduced below some absolute threshold (by definition, the marks per round of a scalable congestion control in steady state should be invariant for any flow rate).

References

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A Approximations

The per-ACK EWMA is not intended to mimic a per-RTT EWMA. Otherwise, the per-ACK EWMA would have to reach the same value by the end of the round, irrespective of whether markings arrived early or late in the round. It is more important for the EWMA to quickly accumulate any markings early in the round than it is to ensure that the EWMA reaches precisely the same value by the end of the round.

Neither is it important that a per-ACK EWMA decays at precisely the same rate as a per-round EWMA (assuming they both use the same gain). The gain is not precisely chosen, so if a per-ACK EWMA decays somewhat more slowly, it is unlikely to be critical to performance (if so, a higher gain value can be configured).

However, it is important that a per-ACK EWMA decays at about the same rate however many ACKs there are per round, although the decay rate does not have to be precisely the same.

The per-ACK approach uses the approximation that one reduction with gain 1/g is roughly equivalent to n repeated reductions with 1/n of the gain. Specifically, that $(1 - 1/ng)^n \approx 1 - 1/g$.

$$(1 - 1/ng)^n = 1 + \frac{n}{-ng} + \frac{n(n-1)}{2(-ng)^2} + \dots$$
$$= 1 - \frac{1}{g} + O\left(\frac{1}{g^2}\right)$$
$$\approx 1 - \frac{1}{g}$$

To get a feel for the size of the error, we define the effective gain (1/g') as the gain of a per-RTT reduction that would be equivalent to multiple smaller per-ACK reductions using the original gain (1/g). Numerically, we find that $g' \approx g + 1/2$.

For instance, multiple reductions with g=16 are roughly equivalent to one reduction with $g'\approx 16.5$; or multiple reductions with g=4 are roughly equivalent to one reduction with $g'\approx 4.5$.

This can be explained because most of the error comes from omission of the $O(1/g^2)$ term. So we set

$$1 - \frac{1}{g'} \approx 1 - \frac{1}{g} + \frac{(n-1)}{2ng^2},$$

which, in the worst case of large n, reduces to

$$g' \approx \frac{2g^2}{(2g-1)}.$$

Then the difference between the reciprocals of the effective and actual gains is

$$g' - g \approx \frac{g}{(2g - 1)}$$

Other than for low values of g this difference is indeed roughly 1/2.

The lowest practical value of g is 2. Multiple reductions using g=2 would be equivalent to a single reduction using $g'\approx 2.7$ (i.e. the error in this worst-case is about 0.7).

Document history

Version	Date	Author	Details of change
00A	07 Nov 2020	Bob Briscoe	First draft.
00B	29 Nov 2020	Bob Briscoe	Added cwnd reduction and increase. Defined reusable function repetitive_div(). Corrected use of cwnd to flight, and distinguished current flight, from flight_ when marks were averaged. Added abstract; schematic of problem; sections on evaluation plan, related work and future work; and appendix on approximations.