Security Analysis of the Micro Transport Protocol with a Misbehaving Receiver

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Introduction

- BitTorrent is the most widely used Peer-to-Peer (P2P) application
- Before December 2008, TCP was its default transfer protocol
- Since this date, uTorrent switched from TCP to the Micro Transport Protocol (uTP) which is based on UDP
What’s the problem with TCP and BitTorrent?

- BitTorrent is in most cases background traffic
- BitTorrent uses multiple connection at once
  - TCP distributes the available bandwidth evenly across connections
  - The more connections one application uses, the larger the share of bandwidth
- Old Solution: Configure bandwidth limit
  - Requires knowledge about the own Internet connection
  - Often lets big head room which is unused
  - Bad configured BitTorrent client can harm other people in the network

⇒ Novel transport protocol based on UDP: Micro Transport Protocol (uTP)
Bittorrent declares war on VoIP, gamers

The next internet meltdown

By Richard Bennett

Posted in Networks, 1st December 2008 12:29 GMT

Gamers, VoIP and video conference users beware. The leading BitTorrent software authors have declared war on you - and any users wanting to wring high performance out of their networks. A key design change in the P2P application promises to make the headaches faced by ISPs so far look like a party game. So what's happened, and why does it matter?

Upset about Bell Canada's system for allocating bandwidth fairly among internet users, the developers of the uTorrent P2P application have decided to make the UDP protocol the default transport protocol for file transfers. BitTorrent implementations have long used UDP to exchange tracker information – the addresses of the computers where files could be found – but the new release uses it in preference to TCP for the actual transfer of files. The implications of this change are enormous.

As BitTorrent implementations follow uTorrent's lead – and they will, since uTorrent is owned by BitTorrent Inc, and is regarded as the canonical implementation – the burden of reducing network load during periods of congestion will shift to the remaining TCP uses, the most important of which are web browsing and video streaming.

By most estimates, P2P accounts for close to half of internet traffic today. When this traffic is immune to congestion control, the remaining half will stumble along at roughly a quarter of the bandwidth it has available today: half the raw bandwidth, used with half efficiency, by 95% of internet users. Oops.

When your internet bandwidth is divided by four, you're going to notice. Even the downloading fiends who haunt the message boards at Broadband Reports can see this, as several have noted [1]:

Is bypassing TCP congestion control a good thing for the users of the network? Why should one persons [sic] non-interactive file sharing generating a dozen to a hundred streams be more important than my interactive VoIP call or gaming experience?

http://www.theregister.co.uk/2008/12/01/richard_bennett_utorrent_udp/print.html
Why the sky isn’t falling

Torrent’s switch to UDP and why the sky isn’t falling

By Ilja de Beijnum | Published 3 years ago

In action movies, when the hero has only minutes before the readout on the bomb counts down to zero, with many miles to drive through a congested city in order to get there in time, he’ll just ignore traffic lights and floor it. Wouldn’t it be great if you could do the same thing on the Internet if your torrents don’t download fast enough? Now you can. Maybe.

In a new alpha version of the popular BitTorrent client µTorrent, uTP has been made the default instead of TCP. uTP sends the actual downloads over the UDP protocol rather than TCP. The existing version 1.8.1 also supports uTP, but it’s not enabled by default. This simple swapping of the transport protocol that sits between the IP layer and the application has gotten our friends over at The Register all worked up: “By most estimates, P2P accounts for close to half of internet traffic today. When this traffic is immune to congestion control, the remaining half will stumble along at roughly a quarter of the bandwidth it has available today: half the raw bandwidth, used with half efficiency, by 95% of internet users. Oops.”

The half of the internet traffic figure is already highly debatable, as services like Hulu and iTunes deliver non-P2P video to ever more users. But the quarter of the bandwidth number is entirely speculation. If µTorrent is indeed eschewing congestion control, however, and this catches on, it could be pretty bad.

The TCP protocol that nearly all applications use—especially the ones sending a lot of data—has a traffic
Micro Transport Protocol (uTP)

- uTP is a reliable transport protocol which makes use of UDP
  - Introduces a new congestion control: Low Extra Delay Background Transport (LEDBAT)

- Similarities to TCP
  - Window based flow control
  - Sequence numbers and ACK numbers

- Differences to TCP
  - Sequence numbers and ACKs refer to packets, not bytes
  - SACK uses a 32 bitmask where each bit represents a packet
  - No congestion control (Slow-start, congestion avoidance, ...) → LEDBAT

- Advantages:
  - Detects foreground traffic and automatically slows down
  - Scales better when sending over many connections at once
Micro Transport Protocol (uTP)

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Traffic

Bandwidth

Time

- uTP traffic
- All Traffic (except uTP)
LEDBAT (Congestion Control)

- LEDBAT uses one way delay as the main congestion measurement
  - Will be compared with previous values (not absolute values)

sender

\[ t_{snd} [\text{Data}] \]

\[ \Delta t [\text{ACK}] \]

\[ \Delta t = t_{rcv} - t_{snd} \]

receiver

- And also:
  - Packet loss (\( \text{max\_window} = \text{max\_window} \times 0.78 \))
  - ACKs during one window

- IETF just tries to standardize LEDBAT (Draft 09³)

³http://tools.ietf.org/id/draft-ietf-ledbat-congestion-09.txt
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Attack Idea 1 (Delay Measurement)

Idea
Lying about the one way delay to induce the sender to send more and more packets into the network

How to do that?

Sender

\[ t_{snd} [\text{Data}] \]

\[ \Delta t [\text{ACK}] \]

Attacker ⚠️

\[ \Delta t = 1 \text{ ms} \]
Code Snippet

```
1  scaled_gain = MAX_CWND_INCREASE_PACKETS_PER_RTT
2       * delay_factor * window_factor;
```

Delay Calculation

\[
\text{delay\_factor} = \frac{100 \text{ ms} - \min(\text{our\_hist})}{100 \text{ ms}}
\]

\[
\text{delay\_factor} = \frac{100 \text{ ms} - 0}{100 \text{ ms}}
\]

\[
\text{delay\_factor} = 1
\]
Bandwidth Usage (100 MiB File Transfer)

- **Without Attack**
- **Delay Attack**

Packets per Second vs. Time in Seconds
Attack Idea 2 (Packet loss)

Idea

We do not inform the sender about packet loss

How to do that?

- As an attacker we’re not interested in data integrity
- Save all packets we got sequentially and not according to their sequence number
Attack Idea 2

Normal packet sorting

Packet Loss

0 1 2 3 4 5 6
Attack Idea 2

Normal packet sorting

Packet Loss

Attack: Packet sorting sequentially

0 1 2 3 4 5 6

0 1 2 3 4 5 6
Bandwidth Usage with Loss Attack

![Graph showing bandwidth usage with different attacks](image)

- **Without Attack**
- **Delay Attack**
- **Lazy Opt-Ack Attack**

**Time in Seconds**

**Packets per Second**

0 20 40 60 80 100 120 140 160 180

0 2000 4000 6000
Attack Idea 3 (Selective Acknowledgment)

Idea

Pretend that we got all packets always via SACK

How to do that?

- Again: As an attacker we’re not interested in data interested
- Set all bits of the SACK bitmask to the value 1
- Replace that code:

```c
1 uint m = 0; // SACK bitmask
```

- with this code

```c
1 uint m = UINT_MAX; // SACK bitmask
```
Bandwidth Usage with SACK Attack

![Graph showing Bandwidth Usage with SACK Attack with various attack types: Without Attack, Delay Attack, Lazy Opt-Ack Attack, and Opt-Ack Attack. The x-axis represents time in seconds ranging from 0 to 180, and the y-axis represents packets per second ranging from 0 to 6,000. The graph illustrates the impact of each attack type on bandwidth usage.]
Average Bandwidth Usage in Mbit/s

- w/o Attack
- Delay Attack
- Lazy Opt-ACK
- Opt-ACK
Can we create Congestion?

- Congestion Experiment
  - Constant 50 Mbit/s UDP stream with iPerf
  - Parallel to this file transfer via uTP
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Results

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<tr>
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Randomly Skipped Packets

Idea
A sender randomly skips packets and checks if the receivers acknowledge those packets.

- Countermeasure only against the lazy opt-ack and opt-ack attack
- This countermeasure will reduce the performance a little bit
Performance Analysis

![Bandwidth vs. Time Graph]

- **uTP without Countermeasure**
- **uTP with Countermeasure**

**Bandwidth in Mbps**

**Time (s)**

- 5
- 10
- 15
- 20
- 25
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Conclusion

The main contributions of our paper are:

- Proposed the following three attacks against the novel uTP:
  - delay attack
  - lazy optimistic attack
  - optimistic attack

- Shown a detail evaluation and shown the severity of those attacks

- Presented a countermeasure against those attacks
Questions?

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Delay Comparison

Packets

Without Attack

With Attack

delay_factor

0

0.85

0.9

0.95

1

1.05
```c
uint m = 0; // SACK bitmask

for (size_t i = 0; i < window; i++) {
    if (inbuf.get(ack_nr + i + 2) != NULL) {
        m |= 1 << i;
    }
}
```
```c
uint m = UINT_MAX; // SACK bitmask

for (size_t i = 0; i < window; i++) {
    if (inbuf.get(ack_nr + i + 2) != NULL) {
        m |= 1 << i;
    }
}
```
Congestion Experiment (w/o Attack)

- Constant UDP Stream
- Normal uTP Transfer

Packets per Second vs Time (s)

- Time: 0s to 70s
- Packets per Second: 0 to 6,000

Graph showing the comparison between Constant UDP Stream and Normal uTP Transfer.
Congestion Experiment (Lazy Opt-ACK)
Congestion Experiment (Opt-ACK Attack)