

A comparison of different TCP/IP and DTN protocols over the D-Star Digital Data mode

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Abstract

This paper examines the performance of the Digital Smart Technologies for Amateur Radio - Digital Data mode with various IP and non-IP based protocols. A throughput comparison was performed between IPv4, IPv6, the DTN Convergence Layer and the NORM Convergence Layer. The experimental results show that the DTN NORM Convergence Layer exhibits better performance than TCP/IP, and appears to perform better over difficult radio links.

I. INTRODUCTION

The Icom Digital Smart Technologies for Amateur Radio (D-STAR [1]) family of transceivers and the use of the D-STAR protocol is becoming more and more an integral part of the toolbox used by Amateur Radio operators for emergency communications activities. The D-Star Digital Data (DD) mode (in the Icom ID-1 transceiver) is of interest as the radio transceiver presents an ethernet interface, and thus any protocol that can be transmitted over ethernet can be sent between any pair of ID-1 transceivers.

In the event of more than two transceivers operating on a single channel, which is likely in an emergency communications scenario, there would likely be a lot of traffic on a channel all in contention for the same bandwidth. This would be necessary in the early stages of an incident until normal communications links were restored.

To the best of the authors knowledge, little work has been done to ascertain the impact on the TCP/IP suite of protocols (or indeed any others) of operating multiple transceivers on a single channel in this type of environment. Previously [2], at the 2009 DCC, some initial results of experiments with DTN and IP networking using Icom ID-1 transceivers in Digital Data mode were presented. These results were preliminary and not experimentally verified.

The approach taken was to firstly test the various protocols in a control scenario to get the best possible realistic throughput, followed by a real point-to-point link (approx 9kms). The rest of this paper is organised as follows: in §2 we briefly explain the D-STAR and DTN concepts, §3 gives an overview of the test scenarios, §4 presents our results, §5 our discussion and §6 our conclusions and future work.

II. BACKGROUND

The authors interest in DTN stems from the potential of DTN to be used to support emergency communications activities, especially where multiple different network types converge i.e. AX.25 [3], D-STAR and the set of 802.11 standards [4] that make up what is commonly referred to as “WiFi”.

In this paper we compare the performance of the TCP/IP protocols, TCP [5]–[7], versus two DTN Convergence Layer implementations namely TCP-CL [8] and NACK-Orientated Reliable Multicast Transport Protocol (NORM) [9].

A. Disruption/delay tolerant networking

Disruption or Delay Tolerant Networking (DTN), is an approach to computer network architecture that seeks to address the technical issues in heterogeneous networks that may lack continuous network connectivity or other extreme environments. Some issues to be addressed include large delay for transmissions resulting from either physical link properties or extended periods of network partitioning, routing capable of operating efficiently with frequently-disconnected, pre-scheduled, or opportunistic link availability, high per-link error rates making end-to-end reliability difficult, heterogeneous underlying network technologies (including non-IP-based internets). The DTN architecture [10] uses in-network or node-level storage to provide an overlay network over various types of network infrastructures. This node-level storage allows application messages (*bundles* in the DTN architecture) to be stored on DTN gateways (or nodes) for arbitrary lengths of time, while waiting for a forward path to become available. This clearly differs from the IP model where IP packets must be forwarded immediately, or dropped. The Delay-Tolerant Networking Research Group (DTNRG)¹ has a reference implementation of the protocol [11] available for experimentation, extension and real-world deployment. See [12] for more information on DTNs.

B. Digital Smart Technologies for Amateur Radio (D-STAR)

Digital Smart Technologies for Amateur Radio, commonly known as D-STAR, is a digital voice and data protocol specification, published in 2001, which was developed as the result of research funded by the Japanese government and managed by the Japan Amateur Radio League [13]. The purpose of the research was to investigate digital technologies for amateur radio. While there are other digital on-air technologies being used by amateurs that have come from other services, D-Star is one of the first on-air and packet-based standards to be widely deployed and sold by a major radio manufacturer that is designed specifically for amateur service use.

The D-STAR system supports two types of digital data streams. The Digital Voice (DV) stream used for example on 430-440 MHz contains both digitised voice (3600 bps including error correction) and digital data (1200 bps). Using a DV radio is like having both a packet link and FM voice operating simultaneously. The Digital Data (DD) stream, used only on 1200MHz, is entirely data with a bit rate of 128k bps. An Ethernet connection is used for high-speed DD D-STAR data.

This work is solely concerned with the Digital Data mode available on the Icom ID-1 transceiver.

III. EXPERIMENTAL NETWORK

Figure 1 shows the experimental network used to measure the system performance. Each node in the network consisted of an Icom ID-1 transceiver and a Linux PC. For D-Star testing, both the DTN reference implementations TCP Convergence Layer (TCP-CL) and the NORM Convergence Layer (NORM-CL) were used to investigate DTN performance. NORM was chosen for examination as previous research [14] suggests that NORM would be suited for use in networks that are bandwidth constrained, or networks that suffer from high levels of packet loss.

Two separate network configurations were examined.

- Control - two radios in close proximity for maximum signal strength/minimum interference.
- Point-to-Point - 9km link.

The Control configuration was investigated with both radios operating indoors in an ideal environment. In both configurations, all routing was configured statically, to avoid routing broadcasts interfering with transfer times.

Figure 1 was configured with Icom ID-1 transceivers at the location of EI7IG, and EI8JA. The aerial at EI7IGs location was a Diamond X-5000 with a Diamond X-7000 at EI8JAs location. Under calm

¹www.dtnrg.org

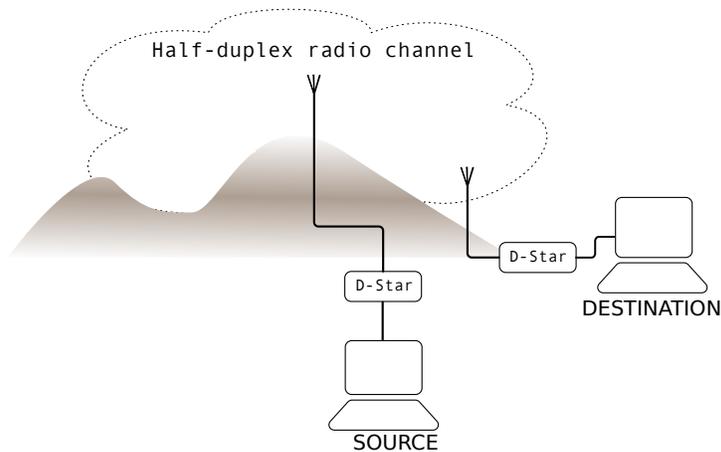


Fig. 1. 9km point-to-point link, not quite line-of-sight

wind conditions, 2 bars of “signal” were visible on the display, with the occasional flicker of a third bar.

The following tests were done in each network configuration:

- IPv4 TCP
- IPv6 TCP
- TCP Convergence Layer
- NORM Convergence Layer

Each test was repeated 25 times to get an average throughput figure for that particular protocol. Care was taken to run the tests under similar atmospheric conditions. The weather station² situated at EI8JA’s location was regularly monitored for wind speed. The iperf [15] tool was used to test TCP on both IPv4 and IPv6. The results for IPv4 were generated with the following command run in a loop 25 times:

```
iperf -c 192.168.254.2 -t 600 -i 10
```

Similarly for IPv6:

```
iperf -V -c fec0::2 -t 600 -i 10
```

Where *192.168.254.2* and *fec0::2* are the IPv4 and IPv6 addresses of the destination node (*192.168.254.1* and *fec0::1* are the source addresses).

The result was a report, with a summary line similar to the following:

```
[ 3]  0.0-601.3 sec  4.90 MBytes  68.3 Kbits/sec
```

To test the DTN Convergence Layers the *dtmsend* utility was used to send a 6MB file across the link. *dtmsend* was configured to ask for a delivery receipt, and the result of this was a report similar to the following:

```
got 33 byte report from [dtn://itx2.dtn]: time=639445.0 ms
```

From these results a spreadsheet was compiled and all results were then converted into bytes per second

IV. RESULTS

NORM was initially configured with a fixed transmission rate of 128kbps. After the initial testing was complete, through experimentation, it was found that configuring NORM for a rate of 84kbps seemed

²<http://aprs.fi/weather/a/EI2WRC>

TABLE I
D-STAR PERFORMANCE - CONTROL CONFIGURATION

Protocol	Min (bytes/sec)	Max (bytes/sec)	Average (bytes/sec)
IPv4	8502.90	8649.84	8576.03
IPv6	5043.07	8372.66	7693.93
TCP-CL	8108.23	8306.87	8242.80
NORM-CL (128)	7850.68	8267.08	8106.88
NORM-CL (84)	9818.84	9926.74	9877.13

to provide the maximum throughput.

TABLE II
D-STAR PERFORMANCE ON POINT-TO-POINT LINK

Protocol	Min (bytes/sec)	Max (bytes/sec)	Average (bytes/sec)
IPv4	85.69	3449.95	1117.19
IPv6	0 †	4268.80	1732.88
TCP-CL	1904.04	7668.61	5527.75
NORM-CL (128)	4440.96	8014.53	6490.19
NORM-CL (84)*	1906.23	8632.93	4362.66

†2 test runs timed out before they could complete; *High winds were experienced on this run — see section V

V. DISCUSSION

Looking at Table I, results seem in line with general expectations. IPv6 achieved less throughput than IPv4, as it has a larger packet header size. The DTN TCP-CL average throughput is between IPv6 and IPv4, i.e. the DTN overhead on an IPv4 packet is less than IPv6 Header overhead. The NORM result is interesting, setting the transmission rate at 84kbps results in an average that is approximately 1300 bytes/sec faster than the next best which is IPv4.

Two of the IPv6 test runs were unable to even start. Both times the TCP connect was sent, no reply was received, this resulted in the connection timing out eventually. All other protocols completed 25 test runs.

Initially, it was planned to include a scenario with a single hop and, also, a node generating interfering traffic. However, when testing began on the point-to-point link it was found that it was far more difficult to achieve a usable, solid connection than prior experience had suggested. Also, a row of trees had grown into the fresnel zone and it was not (legally) possible to have them cut back. This coincided with a period of windy weather, which meant days testing could be completed were few and far between, as the X-7000, being a large vertical antenna, tended to sway in the breeze. Changing to a beam antenna did not help the situation.

Looking at Table II, moving to the “real-world” is even more interesting. The RF path was, in a word, hostile, and this can be seen in the results. IPv4 and IPv6 average throughput reduced significantly. IPv6 has a higher average of the pair, but we are of the opinion that this was due to weather conditions favoring the IPv6 test run (the order of tests was generally IPv4, IPv6, TCP-CL, NORM-CL). A NORM test run with the rate set to 84kbps was run, however wind conditions at the time were far far higher than the others so an attempt was made to stop the script. The TCP session was unable to deal with the poor RF connection and it eventually timed out. The following morning, after the test run completed, it was still impossible to log into the machine remotely over the RF connection to retrieve the logfile. These

results really are a testament to how robust the NORM protocol is for data transfer, under conditions where TCP is unusable.

VI. CONCLUSION

From previous work, we had seen the DTN NORM Convergence Layer showed signs of being more efficient than the TCP/IP protocol over DD mode D-Star radio links. A 15% improvement using NORM over IPv4 is significant enough, what we did not expect to see was a dramatic difference between the robustness of NORM vs TCP.

Looking back at the results, the TCP tests they appear to be broadly in line with what would be expected, in that the throughput is best in IPv4, then TCP-CL, then IPv6. On Icom ID-1 transceivers, NORM appears to have an optimal transmission rate of 84kbps which gives the 15% improvement over IPv4. The robustness of NORM was not something we expected, this became apparent due to the difficult RF path between both nodes and is worth highlighting. The measurements taken provide an illustration that, even without considering robustness in the face of disruption, the ubiquitous TCP/IP protocol is not *always* the best choice.

Future work will include re-location of one of the two point-to-point stations for a more reliable connection. With that in place work can be done on multi-hop transfers and whether DTN performs better in this scenario. At this time the impact and stability of routing protocols in this environment can also be evaluated.

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