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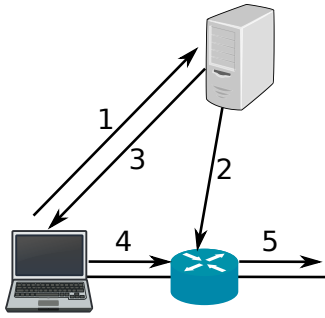


Figure 1: Endhost path query to the controller.

Abstract

Abstract

1 Introduction

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2 Architecture

2.1 Path injections

This paper proposes to integrate TCP with SRv6 to optimise congestion. We reuse a similar controller as in Software Resolved Networks [7]. As shown in Figure 1, an application on the endhost can query (1) the controller for a path to reach a particular destination. The SRN query is implemented by extending DNS requests. The controller computes a path, build its SRH and assign it a binding segment, that we call Path ID. After computation, the mapping between the binding segment and the SRH is inserted in the access router (2). Then, the controller sends the Path-ID inside the DNS reply back to the endhost (3). The application can then build an SRH containing only the Path-ID and insert it in all its packets (4). The access router reads the Path-ID and encapsulate the packet

in an outer IPv6 packet with an SRH that steer the packet through the actual path.

While this first prototype shows that we can build a SDN solution with SRv6, this prototype has two issues. First, it requires to modify applications to talk to a controller. This raises deployment issues if you want to control the path taken by all your applications. Second, it did not consider reaction to congestion, only to failures.

In this paper, we offload the choice of the path to the endhost. The TCP stack of endhosts have access to the congestion state of each connection. This state, along with information from the controller enables the endhost to choose between paths that it received from the controller.

We extend the solution by enabling a daemon, the Path Daemon**TODO: name ?**, in endhosts to request for a set of disjoint paths. The path are disjoint so that we know that a congestion occuring on a path does not influence the others. The endhost can remain agnostic of the actual topology of the network and it will not waste time hitting paths that suffer from congestion on the same link.

We use **TODO: insert algo name ???** from [1] to find these disjoint paths. This algorithm is specific to SRv6 because it compute disjoint paths between two endpoints and checks that these paths can be expressed in fewer segments than a given limit. Some hardware components in their network do not support arbitrarily long SRHs as documented in [?].

The controller computes a set of disjoint paths for every set of pair of source prefix/destination prefix. Each endhost daemon receives a list of Path IDs for pairs matching its access router as source. Along with the Path ID, the controller sends the bandwidth capacity, its current usage and its delay. We scales up the number of controllers with the number of endhost **TODO: inaccurate, it is actually the database but it might be simpler to explain that.**

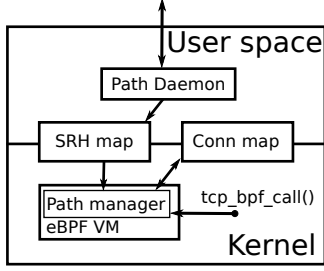


Figure 2: Path management code talks through eBPF maps to the Path Daemon.

2.2 Application-independent path management

The previous section describes the path injection to a daemon that runs on the endhost. However this endhost needs to enable other applications to manage these paths without actually modifying them. A solution could be to modify the kernel directly to include this path management. However, this would require operators to modify the kernel of all the endhosts. An alternative is to inject code in the kernel thanks to eBPF hooks in the kernel. Most of these hooks exists since Linux v.**TODO: version ???**. **TODO: Quid eBPF windows?**

In [7], applications had to be modified to inject their Path IDs in their sockets so that they are inserted in the packets. This limits the deployment of the solution. We need a way to inject the Path ID that is application-independent. Another different with [7] is that we want to react from congestion events inside the endhost. We want to inject a path management behavior inside TCP which is quick to react to congestion in the network. Modifying the kernel directly would require operators to update the kernel of all their machines. Therefore, relying on kernel modification is not the best solution.

Since Linux **TODO: version ???**, programmer can inject code to the kernel thanks to the eBPF VM [?]. We can inject code at various locations, called hooks, each calling a `tcp_bpf_call()` function. We use four of the hooks in our prototype. `BPF_SOCKET_OPS_TCP_CONNECT_CB` is call before sending the SYN packet. This means we can still choose the SRH of the SYN packet in our path management code. We also use `BPF_SOCKET_OPS_STATE_CB` that is called each time the state of TCP changes. We use it to know when the connection is closed to clean the per-connection state that we maintain. `BPF_SOCKET_OPS_RTO_CB` is called after the expiration of the retransmission timer which is an important congestion signal. `BPF_SOCKET_OPS_ECN_CB` is called before sending the segment setting CWR bit in response to packets with ECN bit set. The hook is the only one that we add to the kernel. We choose to place it when sending the CWR instead of the reception of the actual ECN because the CWR bit is sent only once by RTT. This is better done there than in the

path management code.

Figure 2 shows the interactions of the different components inside the endhost. The Path manager code runs in an the isolated eBPF VM inside the kernel space. A verifier [?], packaged with the kernel, checks memory calls and program termination of the code before injection it so that we are guaranteed that the code does not make the kernel crash **TODO: Comments on other verifications ?**. For injected code in the TCP stack, the injected code receives a structure containing the code of the hook that triggered the code, the five tuple, and some variables of the TCP connection, such as the minimum RTT or the congestion window. Moreover, it can read and writes to memory chunks, called eBPF maps, that can be written to or read by a user space application. In this architecture, our daemon will fill a first eBPF map, the SRH map with the Path IDs received from the controller. This map is a hashmap that maps the destination prefix to a list of a structures containing a Path ID, the path bandwidth, its latency,... (i.e., all the pieces of information transmitted by the controller). **TODO: Discuss how the eBPF map can support multiple size of prefixes?** Since eBPF injected code does not have global variable or heap, we use a second eBPF map also oragnized as hashmap, the connection map. The key is the five tuple and it maps to a structure containing data about the connection.

TODO: Explain cgroups?

```

1 int handle_sockop(struct bpf_sock_ops *skops)
2 {
3     struct five_tuple five_tuple;
4     get_five_tuple(&five_tuple, skops);
5
6     struct conn *conn = bpf_map_lookup_elem(&
7     conn_map, &five_tuple);
8     if (!conn) { // New connection
9         struct connection new_conn;
10        memset(new_conn, 0, sizeof(new_conn));
11        new_conn.srh = get_best_path(srh_map, &
12        five_tuple.remote);
13        bpf_setsockopt(skops, SOL_IPV6, IPV6_RTHDR
14        , new_conn.srh, sizeof(*new_conn.srh));
15        init_exponential_backoff(&new_conn);
16        bpf_map_update_elem(conn_map, &five_tuple,
17        &new_conn, BPF_ANY);
18        *conn = new_conn;
19    }
20
21    switch (skops->op) {
22        case BPF_SOCKET_OPS_STATE_CB:
23            if (skops->args[1] == BPF_TCP_CLOSE)
24                bpf_map_delete_elem(conn_map, &
25                five_tuple);
26            break;
27        case BPF_SOCKET_OPS_ECN_CB:
28            // TCP segment with CWR bit set
29            react_to_congestion(five_tuple, conn);
30            break;
31        case BPF_SOCKET_OPS_RTO_CB:
32            // Retransmission timer expiration
33            react_to_congestion(five_tuple, conn);
34            break;
35    }
36    socks->reply = rv;

```

```

32     return 0;
33 }

```

Listing 1: Path management

Listing 1 shows the pseudo code of the path management. When an application starts a connection to a server, the eBPF hook ??? is triggered. We start by retrieving the five tuple and check if an entry for this connection exists. Since it is the new connection, it won't find it. Therefore, we create this entry, choose the best path for the destination and change set the SRH with the Path ID in the socket so that it is inserted in every packet sent. We also instantiate timers to apply an exponential backoff to choose when we modify the path **TODO: too soon in the text ?**. After that, we identify the hook that triggered the call. When the TCP state changes, and when the connection is closed, we clean the connection state of the eBPF map. When we receive a congestion signal, either triggered by ECN or by the retransmission timer expiration, we call another routine whose pseudo code is given in Listing 2.

In this routine, we first check that we waited a sufficient amount of time before moving the connection. Indeed, we implement an exponential backoff to prevent stability issues. Without this, all connections in the network would switch paths at the same time and this might cause congestion in another part of the network. Then, they would all switch back to previous situation and continue flapping **TODO: check term** between both paths.

Then we select the path with the highest available bandwidth. This information originates from the controller and its monitoring of the network through, e.g., OSPF-TE [?], ??? or ???. **TODO: Do we monitor the actual bandwidth of the path we are currently using?** If a better path was found, we switch path by setting the SRH to the socket and update the timers for the exponential backoff.

```

1 void react_to_congestion() {
2     // Wait exponential backoff
3     if (current_time - conn->last_move_time < conn
        ->wait_before_move)
4         return;
5     // Select the path with the highest bandwidth
6     for (i = 0; i < MAX_SRH_BY_DEST; i++) {
7         srh_entry = &dst_infos->srhs[i];
8         current_bw = srh_entry->bw;
9         if (current_bw > highest_bw) {
10             best_srh = srh_entry->srh;
11             highest_bw = current_bw;
12         }
13     }
14     if (best_srh != null) {
15         bpf_setsockopt(skops, SOL_IPV6, IPV6_RTHDR
            , best_srh, sizeof(*best_srh));
16         conn->srh = best_srh;
17         update_backoff_timers(conn);
18     }
19 }

```

Listing 2: Reaction to congestion

TODO: Listings for each portion of code

3 Evaluation

We used 233 topologies from the Topology Zoo [6]. We modified these topologies to remove part of the graphs that lack link redundancy, . . . trees. The resulting topologies have only routers with at least two links. Communication between routers in a tree can only use a single path. Such trees cannot see any improvement from using Segment Routing. Moreover, such trees are likely artifacts of the topology collection because they cannot preserve connectivity even with a single link failure. Figure 3a and 3b shows the CDF of the number of routers and links in the resulting topologies. Figure 3c shows the distributions of routers percentage with a given number of attached links.

The set of topologies does not include which routers are actually linked to endhosts and which are core routers. We assume that routers with two links are edge routers and the others are core routers because they are more connected.

3.1 Maximum improvement

This sections evaluation the theoretical improvement of using multipath transport protocols and Segment Routing on the endhosts instead of a single path protocol with only shortest paths.

We used a linear programming formulation of the Max Flow problem. Given a set of paths linking a set of the edge routers, we compute the maximum amount of traffic that can go through the topology without exceeding the link capacity.

$$\begin{aligned}
 & \max_f \sum_{p \in \mathcal{P}} f_p \\
 & \text{s.t.} \quad \sum_{p \in \mathcal{P}} r(l, p) \cdot f_p \leq c(l) \quad \forall l \in \mathcal{L}
 \end{aligned}$$

\mathcal{P} represents the set of sr-paths, encoded as a list of intermediate routers. As for IPv6 SR, The real path taken between each intermediate router is the shortest one. $r(l, p)$ represents the ratio of flow that goes through a particular link. This ratio is used to model the ECMP splitting of the traffic.

To model a single-path transport protocol without SR we insert all the shortest sr-paths between each pair of edge router to \mathcal{P} . We assume that we have a high number of connections for each path and therefore that hash-based ECMP splits the traffic evenly. [2] shows that it is not the case in practise.

Multipath transport protocols with SR also uses shortest path routing. However, they can establish a big number of subflows for each connection and hope that they will cover all or a portion of the shortest paths. This hijacks ECMP that sees subflows as independent connections. Flow Bender [5] and SEMPER [4] used this solution to leverage some of the

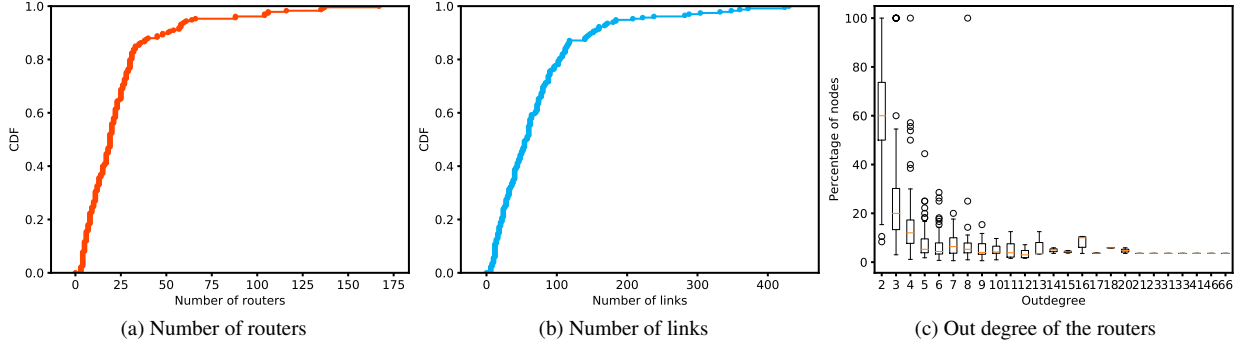


Figure 3: Features of the evaluation topologies

path diversity of a topology. We will assume for simplicity that this method works perfectly and that multipath protocols can always cover all the shortest paths. We insert a sr-path for each shortest path between each pair of edge routers instead of giving one sr-path representing all the shortest paths between a given pair.

Introducing Segment Routing means adding more paths to \mathcal{P} . Note that generating the set of all possible paths is in $O(|\mathcal{R}|^k)$ with \mathcal{R} being the set of routers and k the maximum number of intermediate nodes. This is not practicle for big topologies. Therefore, we reuse the approach of [?] that iteratively adds relevant paths to the set until no interesting paths can be added to \mathcal{P} . When it reaches this point, computing the model on this restricted set or on the complete one gives the same solution. **TODO: Explain how the model is different from infocom, i.e., demands become unbounded or is it too detailed ?**

Adding IPv6 SR can significantly improve the maximum throughput. Figure 4a shows the relative maximum flow improvement of using single path or multiple path with SR. There is a boxplot for a given percentage of edge router pairs that have heavy hitters to route **TODO: Give the rational behind the choice of the values.** The number of heavy hitters has influence on the relative improvement. It is easier to unused alternative paths with fewer heavy hitters. With 1% of heavy hitter pairs, for half of the topologies, we can improve the maximum flow by 71%. For 10% of heavy hitter pairs, we can still improve it by 30%. Figure 4b shows the improvement with multiple path protocol without SR. The improvement is about 5% lower. This justifies the use of SR even with multipath protocols.

3.2 Largest set of disjoint paths

Finding the largest set of disjoint paths between two routers can be done with Edmonds–Karp algorithm [3] in $O(|\mathcal{R}||\mathcal{L}|^2)$. Even faster strategies can imply applying Dijkstra algorithm iteratively. Each time that the algorithm finds a path, we remove its links for the next computations. We stop it

when it cannot find another path. This is solved in $O((|\mathcal{L}| + |\mathcal{R}|)n \log |\mathcal{R}|)$ with n being the number of iterations of dijkstra algorithm. $O(n)$ cannot be bigger than $O(|\mathcal{L}|)$ because at each iteration, if a path is found, the links if this path will be removed. In practise, source and destinations are behind routers with exactly two outgoing links. There will be at most two disjoint paths. Therefore, applying the Dijkstra algorithm iteratively is faster than using the Edmonds-Karp.

Figure 5a shows a CDF of the mean number of disjoint paths by pair of access router in each topology. We consider routers with exactly two links as access routers. This means that there are at least one path between two routers but at most two since there are only two outgoing links. Edmonds-Karp computes the largest set of disjoint paths that can be reached. We can reach the same sets for 70% of topologies by applying iteratively Dijkstra algorithm.

SRv6 is limited in the number of segments that it can apply in a packet. This limit can be due to hardware limitationsi of IPv6 SR implementations [8]. So we need to use algorithms that prevent the number of segments to grow beyond a given limit. Always finding the largest set of disjoint paths within the limit of segment is exponential and therefore not feasible in practise. [1] defines the **SPSTODO: Check with Francois for the name** algorithm that computes the shortest path between two routers under a given limit of segments. Each time that the algorithm finds a path, they remove all the links for the next computations. They stop when no more path can be found. Figure 5b shows that the number of disjoint paths increases with the number of authorized segments. With only one segment, we can only have one path because only the destination can be encoded in the segment list. Increasing to 4 segments, (i.e., 3 intermediate routers), signigicantly improves the number of disjoint paths. We see that increasing the limit beyond this does not improve significantly the number of disjoint paths.

TODO: Conclusion

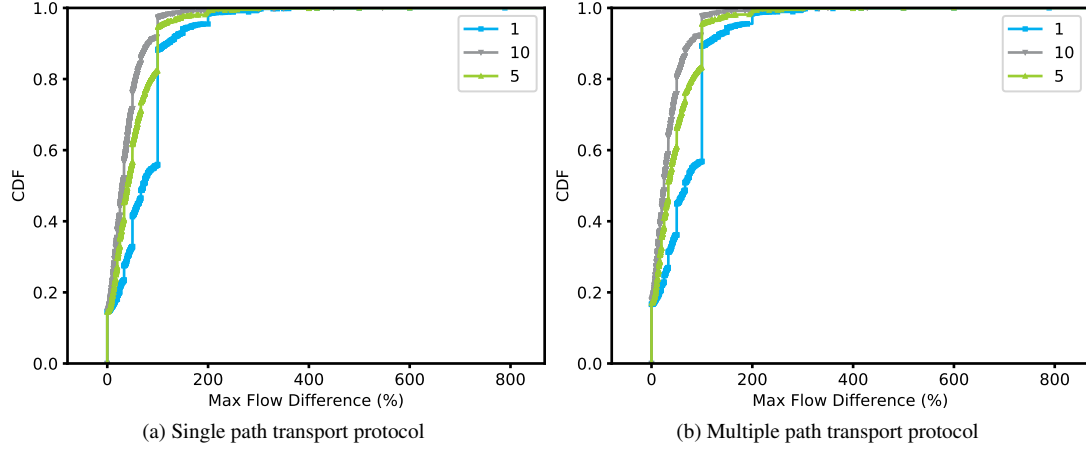


Figure 4: Relative Maxflow improvement of adding IPv6 SR

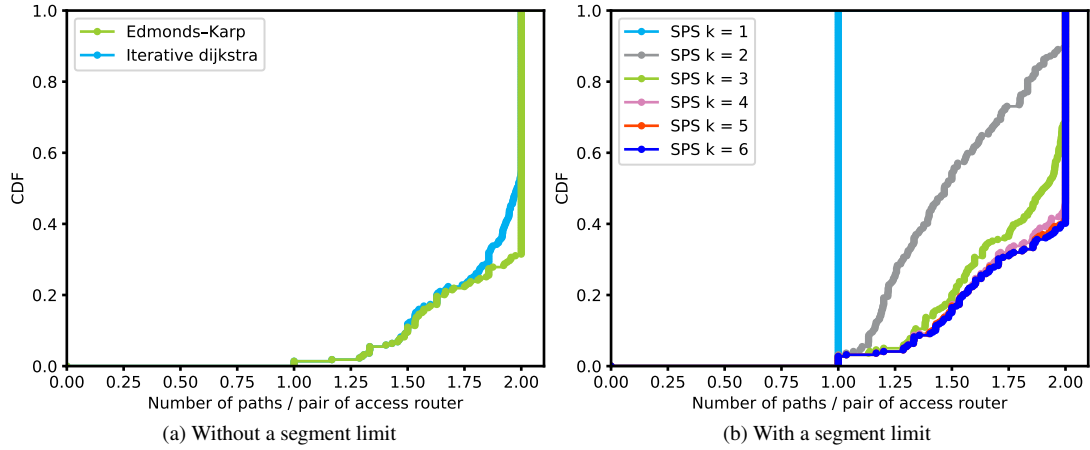


Figure 5: Number of disjoint paths found by pair of access routers

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