Responsive Multipath TCP in SDN-based Datacenters and Dynamic Scaling of Virtual Network Functions

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Overview

- First Section
 - Responsive Multipath TCP in SDN-based Datacenters

- Second Section
 - Dynamic Scaling of Virtual Network Functions

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Responsive Multipath TCP in SDN-based Datacenters

• Section 1: Responsive Multipath TCP in SDN-based Datacenters.

Background and Motivation

- In datacenter network, content replication, virtual machine migration, and data shuffling in MapReduce tasks generate many large flows.
- These large flows constitute the majority of datacenter traffic.
- Efficient transfer of these large flows is crucial to the performance of a datacenter network.
- But current design of both transport layer protocol and routing scheme is not efficient enough in supporting the high throughput of large flows.

Routing in Datacenter Network

 Datacenter network has redundant paths that connect servers from different racks.

• How to route?

Routing in Datacenter Network

- Rely on ECMP to balance the traffic on different paths.
- But ECMP is not intelligent.
- Two large flows may be routed on the same paths, so none of them can achieve optimal throughput.

Routing in Datacenter Network

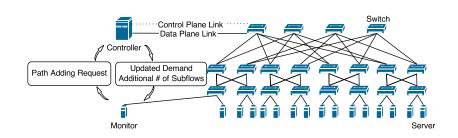
- Calculate route using SDN controller, like Hedera.
- More intelligent, can avoid paths collision, constantly re-balance entire traffic.
- But controller needs to constantly polls traffic statistics from all switches. Bad scalability and responsiveness.
- Existing flows may be re-routed to another path. Possible packet reordering and packet loss.

- Traditional TCP has been shown to be inefficient in datacenter network.
- People design TCP variants that target the needs of datacenter network, such as DCTCP and D2TCP.
- Their limitation is that they are not multi-path protocol.
- Failed to fully utilize path diversity in datacenter network.

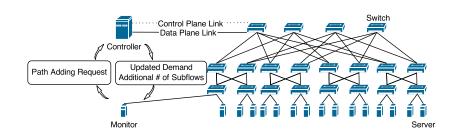
- Use multi-path based transportation protocol to exploit available bandwidth.
- Split one TCP flow into multiple subflows.
- So that available bandwidth is more efficiently used.
- How to split?

- Split at the switch.
- Reply on advanced switch features.
- Hard to deploy in a large scale.

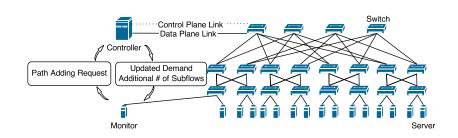
- Split at server, using MPTCP.
- Then flows are still routed using ECMP. We still have all mentioned potential problems.
- Number of subflows used by each flow is fixed. Add unnecessary overhead, can't react to traffic conditions.



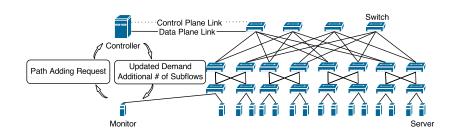
- Responsive MPTCP system for SDN-based datacenters.
- Controller + Monitor design.



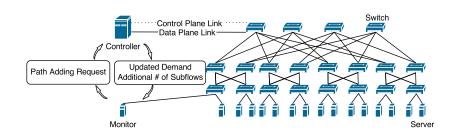
- Controller estimates the demand for each MPTCP flow.
- Controller sends updated demand to the monitor.



- Monitor records updated demand and current throughput.
- Monitor issues path adding request when necessary.



- Controller calculates needed additional subflows.
- Controller sends this information back to monitor.



- Dynamically decide the number of subflows to be used by each MPTCP flow and the best subflow path.
- With this system, better throughput, smaller overhead.

Controller

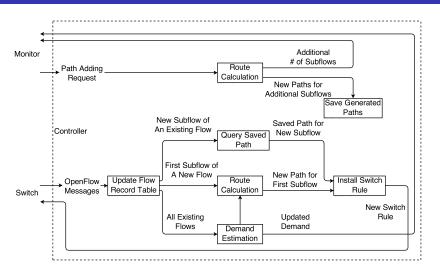


Figure: Architecture of Controller

Controller: Demand Estimation

- Fat-tree datacenter network is a fully non-blocking network.
- Bandwidth bottleneck along a flows path is either senders access link or receivers access link.
- We employ the demand estimation algorithm in Hedera to calculate flow's demand.
 - Proportionally increase the sending rate at sender's access link.
 - Proportionally decrease excessive sending rate at receiver's access link.
 - Stop until all sending rates stablize.

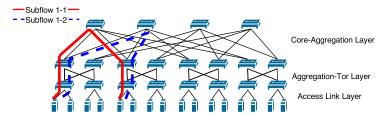
Controller: Demand Estimation

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Algorithm: Demand Estimation and Dispatching
Input: demand record d_{-}t, new flow f_{-}n or expired flow f_{-}e
1: old_-d_-t \leftarrow d_-t:
2: update d_{-}t to include f_{-}n or exclude f_{-}e;
3: run Hedera demand estimation algorithm with d_{-}t as input;
4: for each flow f in d t do
5:
        if f is not in old d t then
6:
            dispatch d_{-}t[f] to the monitor at sender of f;
7:
        else
8:
            if |d_t[f] - old_d[f]| > \delta_{DDT} * old_d[f] then
9:
                dispatch d_{-}t[f] to the monitor at sender of f;
10:
            end if
11:
        end if
12: end for
```

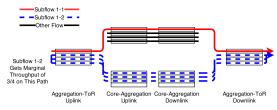
- Route calculation algorithm computes a set of new paths for the flow.
- It uses demand gap as input.
- The goal of route calculation algorithm is to cover as much the demand gap as possible.

- Property 1: Aggregate MPTCP Flows Fairly Share Link Bandwidth.
 - When multiple subflows of a MPTCP flow traverse the same link, they are viewed as one aggregate flow.
 - Aggregate MPTCP flows fairly share the link bandwidth, because of MPTCP congestion control protocol.
 - So when n aggregate flows are contending for a link with bandwidth B, the expected throughput of each aggregate flow can be calculated as B/n.

- Property 2: Marginal Throughput.
 - Subflows at most Share Links in the Aggregation-ToR Layer, as illustrated in the following figure.



- Property 2: Marginal Throughput.
 - Subflows at most Share Links in the Aggregation-ToR Layer.
 - Marginal throughput quantifies the achievable throughput of a subflow on a new path.
 - Especially when new subflow and existing subflows share links on this new path.



- We apply the following two rules to calculate the expected throughput of a new subflow sf of flow f on a given path p.
- **Rule 1**: If there is no other subflow of f that uses p's aggregation-ToR uplink, find out the largest number of aggregate flows sharing the same link on p, n_{max} . The expected throughput of sf on p is B/n_{max} .
- Rule 2: If there are m existing subflows of f that use p's aggregation-ToR uplink, calculate p's marginal throughput m_-t , as well as the upper bound of sf's expected throughput B/n_{max} .

Controller: Demand Estimation

Algorithm: Route Calculation **Input**: flow f, demand gap d_{-g} , number of existing subflows $n_{-s}f$ 1: set number of additional subflows $n_new_sf = 0$; 2: while $n_new_sf + n_sf < M$ do 3: find out a subflow path p with the largest expected throughput e_-t ; 4: if f is a new flow then 5: install switch rules on p: 6: return; 7: end if 8: if e t == 0 then 9: return; 10: end if 11: save path p; 12: $n_{-}new_{-}sf + = 1$: 13: if $e_-t > d_-g$ then 14: notify the sender monitor of f of n_new_sf ; 15: return: 16: else 17: $d_{-}g = d_{-}g - e_{-}t$; 18: end if 19: end while 20: notify the sender monitor of f of n_new_sf; 21: return:

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Monitor

- Monitor is a daemon program running on each server.
- It keeps track of current throughput of each active flow.
- It constantly receives updated demand from controller.
- It issues path-adding request when it detects a significant gap between current throughput and updated demand.
- Monitor daemon is periodically executed every t_m seconds.

```
Algorithm: Monitor Loop
Input: monitored flows f[n], previous flow rates p_{-}r[n], counters count[n], flow demand
demand[n]
1: for i = 1 : n do
2:
        obtain instant throughput i_{-r} for flow f[i];
3:
      p_{-}r[i] = 0.2 * p_{-}r[i] + 0.8 * i_{-}r;
4:
       if |i_{-}r - p_{-}r[i]| < \delta_{RVT} * p_{-}r[i] then
5:
            count[i] = count[i] + 1;
6:
            if count[i] == R then
7:
                count[i] = 0;
8:
                if p_{-}r[i] < (1 - \delta_{DGT}) * demand[i] then
9:
                    gap = demand[i] - p_r[i];
10:
                    issue path adding request for f[i] with gap;
11:
                    return:
12:
                end if
13:
            else
14:
                return;
15:
            end if
16:
        else
17:
            count[i] = 0;
18:
            return:
19:
        end if
```

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20: end for

- We evaluate our responsive MPTCP system using a NS3 simulator.
- We simulate a datacenter network with a 8-Ary fat-tree topology which contains 128 servers connected using 1Gbps links.
- Up to 4 subflows per MPTCP connection is allowed.
- Performance under different parameter settings is evaluated.
- Comparison against ECMP and Hedera is conducted.

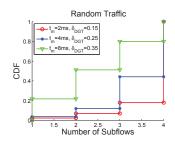
TABLE I: Summary of Experiment Results

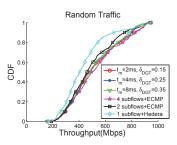
	Random Traffic		Permutation Traffic		Shuffling Traffic	
	NS	TP(Mbps)	NS	TP(Mbps)	JCT(ms)	NS
$t_m = 2ms, \delta_{DGT} = 0.15$	3.72	521	3.88	766	1402	1.92
$t_m = 4ms, \delta_{DGT} = 0.25$	3.40	513	3.82	767	1441	1.28
$t_m = 8ms, \delta_{DGT} = 0.35$	2.47	506	3.75	760	1454	1.03
4 subflows+ECMP	4	530	4	734	1332	4
2 subflows+ECMP	2	493	2	643	1394	2
1 subflow+Hedera	1	438	1	747	1652	1

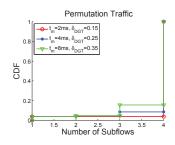
NS: Average number of subflows per flow.

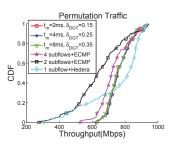
TP: Average throughput per flow.

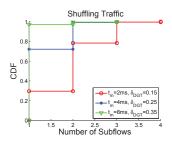
JCT: Average shuffle completion time per MapReduce job.

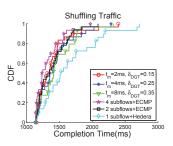












Dynamic Scaling of Virtual Network Functions

• Section 2: Dynamic Scaling of Virtual Network Functions.

Background

- Network middleboxes, i.e. NAT, Firewall, Proxy, are running everywhere.
- Many of them are usually implemented as proprietary hardware, making them notoriously hard to upgrade and scale.
- With cloud and virtualization technology, network middleboxes with a software-implementation can be run on virtual machines.

Background

- Efforts have been made on improving the performance of NFV system:
 - Increase the processing speed of software network middlebox running on virtual machine.
 - Dynamically scale network middleboxes in face of traffic change.

IMS System

- We start our research by studying a specific network middlebox system: Ip Multi-media Subsystem (IMS).
- We use an open-source implementation of IMS system, Project Clearwater.

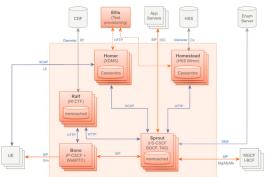
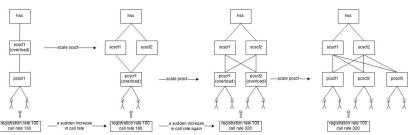


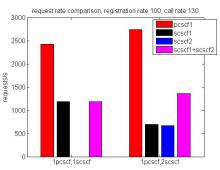
Figure: Architecture of IMS System, from Project Clearwater Website

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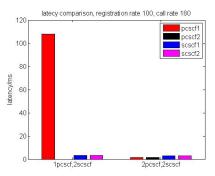
- In order to dynamically scale the system, we need to identify the bottleneck in the system.
- We design the following simple experiment to show the bottleneck of the system.



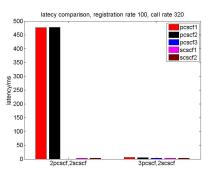
- With registration rate 100 and call rate 130, SCSCF overloads.
- SCSCF performs blocking operations that retrieve user information from database.
- When it overloads, it fails to serve some calls and for some calls to terminate.



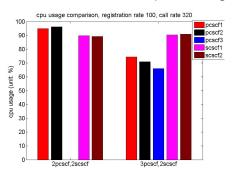
- With registration rate 100 and call rate 180, PCSCF overloads.
- PCSCF performs non-blocking operations, acting as a pure proxy.
- When it overloads, its request processing latency increases drastically.



- With registration rate 100 and call rate 320, PCSCF overloads again.
- SCSCF will overload soon, if workload keep increasing.



- With registration rate 100 and call rate 320, PCSCF overloads again.
- SCSCF will overload soon, if workload keep increasing.



What We Learn

- Under different workload, different part of the system becomes bottleneck.
- It's hard to design an accurate mathematical model that takes workload as input and output the number of required resources.

What We Plan to Do

- Monitor the overloading signal. (Latency, CPU usage, Successfully Completed Requests)
- Scale the system in react to the overloading signal.
- After scaling, need to balance the load on each instance.
- Scale up vs scale down:
 - It's easier to scale up than to scale down.
 - Because it's even harder to determine whether you can scale down.

The End

• Future Work:

- Along the direction of datacenter network, design more efficient datacenter transport protocols with control plane assistance and implement them in real system.
- Along the direction of NFV, carry on this existing research and find out more interesting from it.

The End

• Thank you and Q & A!