

Resilient Advertisements

Paper #559, 12 pages body, 19 pages total

Abstract

Abstraction

1 Introduction

Cloud and content providers (hereafter Service Providers) enable Internet applications used daily by billions of users. The applications have increasingly stringent performance and reliability demands, as the Internet is increasingly used for mission-critical applications (*e.g.*, enterprise services [?]) and as performance requirements become tighter (*e.g.*, virtual reality requires **TBD: X** ms latency and **TBD: Y** ms jitter [?]). To meet these stricter requirements with the same fundamental Internet protocols, Service Providers have deployed interconnected sites and connected with thousands of networks in hundreds of locations, offering rich connectivity and distributed capacity.

The deployment’s physical resources (*i.e.*, peering links, servers) are provisioned to accommodate demands from (ingress) and to (egress) user networks. Service Providers balance egress demands across links to minimize congestion and latency [12, 15, 5]. However, it is difficult to control exactly how much traffic reaches each site and how much *ingresses* each link into the Service Provider’s network since (a) ingress traffic can only be controlled indirectly, by directing users to different *prefixes*, and (b) it is difficult to advertise prefixes in a way that balances load in dynamic network conditions.

Given this lack of control, resource overload occurs due to link failures [9], DDoS attacks [? ? ? ?], flash crowds [?], and route changes that cause significant traffic shifts [? ? ? ? 9]. This overload leads to degraded service for user traffic [? ? ?]. Service Providers may respond to resource congestion by (a) overprovisioning resources [? ?] or (b) draining congested links and moving some of that traffic to other resources with free capacity [9]. We demonstrate in ?? that overprovisioning resources can require **TBD: X%** more capacity than average load, significantly increasing expenses.

Recent work drains traffic from overloaded links/sites by

withdrawing prefix announcements that are also advertised via other (healthy) links/sites [? ?], so that traffic destined to those prefixes is forced to arrive on other links/sites after BGP converges (within tens of seconds [?]). However, this solution to overload is post-hoc, unpredictable, and ineffective. Other solutions use traffic engineering systems in user networks to optimize latency and dynamically shift traffic [?], but do not provide guarantees that good backup options exist. **TBD: We demonstrate** both of these options fail to provide enough deployment resilience to handle even moderate variability and traffic increases in ??.

To solve these problems, we propose a system — SCULPTOR— that proactively advertises multiple prefixes to peers to expose diverse routing options, then assigns user traffic to paths towards these prefixes to optimize latency and minimize overload during both steady-state and failure. That is, SCULPTOR ensures that advertisements are resilient to changes in traffic distributions and failed sites/peering links. Our key insight is to frame this problem as finding a way of advertising reachability that maximizes latency subject to capacity constraints both with and without failures, as congestion can be viewed as a partial deployment failure. Our solution leverages the rich connectivity of Service Providers, balancing traffic over backup paths through many sites and links as conditions change. We solve different research challenges than in the intradomain routing fault-tolerant-networking literature [? ? ? ?] since, unlike in that domain, we can neither control nor predict routes perfectly from users to Service Providers (interdomain routes are determined by BGP and thus by intermediate organizations). Instead, we build a model of how users are routed to Service Providers, and iteratively refine this model using measurements while using the model to optimize advertisement strategies using gradient descent.

We make two contributions. First, we present an optimization framework that can be used to minimize performance metrics (*e.g.*, maximum link utilization, latency, latency under single link failures) over different sets of prefix advertisements. Part of our framework is a model for predicting performance in unknown scenarios—we compute probability

distributions of performance metrics in advertisement configurations we have not measured, which is important as changing routing configurations and then issuing measurements is slow. This model enables SCULPTOR to efficiently optimize over a large space without measuring every possible configuration. We demonstrate that this model quickly learns how to accurately predict performance metrics using few measurements compared to alternate approaches. We then optimize these (modeled) performance metrics using gradient descent which performs particularly well in our setting due to the high dimensionality of the problem and high degree of parallelism gradient descent admits.

Second, we prototype and evaluate our framework in a system, SCULPTOR, at Internet scale using the PEERING testbed [11], which is now deployed at 32 Vultr cloud locations [14]. Vultr is a global public cloud that allows us to issue BGP advertisements via more than 10,000 peerings. We demonstrate that SCULPTOR effectively computes prefix advertisements that give users low latency both during normal operation and variable network conditions.

SCULPTOR is able to reduce median latency under single link failures by **TBD: X** ms compared to the state of the art. We model SCULPTOR’s impact under a variety of ingress link capacities (*i.e.*, having more available capacity means there are fewer problems to solve, but comes with more cost). We demonstrate that SCULPTOR reduces necessary ingress capacity by **TBD: Y** times compared to the state of the art while retaining **TBD: Z** percent of the benefit. Hence, SCULPTOR serves both as an optimization tool for current deployments and as a network-planning tool—informing us where capacity is and is not needed.

2 Motivation and Key Challenges

2.1 Setting

Service Providers offer their services from tens or hundreds of geo-distributed sites. Each site offers identical service so can serve any user, but users want to reach a low-latency site for performance. Sites consist of sets of servers which have an aggregate capacity. Service Providers also connect to other networks (often thousands) at sites via dedicated links or shared IXP fabrics. Each such link also has a capacity. When utilization of a site or link (collectively, a resource) nears/exceeds 100%, performance suffers, so Service Providers strive to keep utilization reasonably low. Resources can also fail completely due to, for example, physical failure and misconfiguration.

Ensuring users can reliably route over healthy paths to Service Providers even during partial system failure is increasingly important as shown by the considerable attention that related problems have seen in the news and research community. Microsoft recently stressed the importance of resolving congestion on ingress links [9], developed a system to identify performance problems on paths [?], and is consider-

ing deployment deep within user networks to further enhance ingress routing performance and reliability [?]. Google similarly developed a system to identify performance problems on paths [7]. Peering disputes regularly make their way into the news, and these disputes often lead to long-standing congestion on interdomain links **TBD: cite**. DDoS attacks are an ever-present problem, as shown by recent events **TBD: cite recent events or work** and by several publications presenting new methods of mitigating DDoS attack effects **TBD: cite**. Research from Facebook, Google, and Microsoft all demonstrates considerable recent focus on providing service even under partial system failure (as partial system failure is a constant in a large networked system) [? ? ? ?].

Moreover, new trends in service offerings and application usage make operation under dynamic conditions both more challenging and important. Service Providers increasingly offer mission-critical services such as enterprise solutions [? ? ? ? ?], so ensuring performant, reliable operation is more important for these services now than ever. Recent work shows that user traffic demands are highly variable due to flash crowds, congestion, and path changes and so are much harder to plan/optimize for than inter-datacenter demands [?]. Operators regularly report such events on social media and blog posts **TBD: cite**. Such trends could become more salient as 5G/next-generation applications drive massive amounts of traffic from users to services [10], making user demands even more variable and more challenging to satisfy.

2.2 Approaches to Interdomain Routing

A challenge in offering low-latency, reliable services to users is routing traffic from user networks to Service Provider networks since Service Providers lack full control of which interdomain path traffic takes. BGP, the Internet’s interdomain routing protocol, computes paths in a distributed fashion, giving each intermediate network a say in which paths are chosen and which are communicated to other networks. Service Providers can, however, advertise their reachability to peers/providers in different ways to increase the chance of there being good paths for users. Today, Service Providers either direct users to specific sites using unicast prefix advertisements [7, 12?], or use anycast prefix advertisements to provide relatively low latency and high availability at the expense of some control [1, 6?] **TBD: more**.

anycast, where Service Providers advertise a single prefix to all peers/providers at all sites, offers high availability following failures since BGP automatically ensures reachability to the deployment after tens of seconds for most networks [?]. Prior work shows that this availability comes with higher latency in some cases [8, 6]. unicast advertises a unique prefix at each site, enabling user redirection to a particular site [3]. Hence, recent work proposes hybrids of unicast and anycast which achieve a happy medium between performance and availability [1, 16? ?]. This latter

class of solutions advertises different prefixes to subsets of all peers/providers to offer users many low-latency options across different links/sites. We refer to these solutions as *selectivecast* solutions since they are *selective* about who they advertise prefix reachability to, and have traits of both anycast (advertising at many sites to many peers/providers) and unicast (many prefixes, selectivity).

2.3 Current Approaches Do Not Consider Dynamic Conditions

In anycast, unicast, and *selectivecast* scenarios, Service Providers set capacities based on steady-state demand. For example, Service Providers might overprovision resources by a fixed percentage. Hence, none of these approaches to advertising reachability (unicast, anycast, or *selectivecast*) explicitly plan for dynamic traffic conditions. For example, AnyOpt and PAINTER (*selectivecast* solutions) find low-latency paths [16?], but it is unclear whether these paths can satisfy user demands under a flash crowd, and it is unclear how to scale these approaches to account for such scenarios. None of these approaches *even take user demands into account* when computing how to advertise reachability.

Instead, the current state-of-practice to handle dynamic conditions is to drain traffic from overloaded links/sites by *withdrawing* prefix announcements that are also advertised via other (healthy) links/sites [? 9?]. The traffic destined to those prefixes then arrives on other links/sites advertising those prefixes after BGP reconverges. However, this solution to handling dynamic conditions is post-hoc, unpredictable (as BGP is unpredictable), and ineffective. For example, TIPSy only achieves 70% prediction accuracy and (in a case study) required several prefix withdrawals to mitigate a single congested link [9].

Another common solution to tackling dynamic conditions is to overprovision resources to handle predicted peak loads in the future [? ?], but Figure 1 demonstrates that doing so can incur excessive costs. Figure 1 computes differences in peak utilization on links between different successive time periods, using longitudinal link utilization data from OVH cloud [?]. We split the dataset into successive, non-overlapping N -day periods and compute changes in 95th percentile link utilization from one period to the next. We choose this percentile to reduce noise. Even when computing near-peak loads over successive 120-day time periods, near-peak loads can increase by 10% for 7% of links, 20% for 2% of links, and increase as high as 70%, which illustrates the amount Service Providers must overprovision to ensure they can meet resource demands. Using backup paths to handle such (rare) events can help keep costs low.



Figure 1: Planning for future peak loads requires excessive overprovisioning.

2.4 Key Problem

The key problem with current practices of routing users to Service Providers is that Service Providers have no guarantee that backup paths/sites can handle new traffic loads. Prior work finds ways to prepare for intradomain failure even under dynamic traffic conditions [? ? ? ? ?], because in that setting the network has full *control* of how traffic flows over paths and so can ensure demand does not exceed capacity.

Compared to this intradomain setting, two key differences in the interdomain setting are that (a) the network cannot precisely control the paths traffic takes and (b) the network cannot easily determine the bottleneck capacity of paths before placing traffic on them. We explicitly address the former problem and leave the latter problem as future work. That is, we consider the problem of ensuring reliable (failure-resistant) interdomain routing when we know the bottleneck capacity of backup paths. We approximate the capacity of a path as the capacity of the corresponding peering link through which that path ingresses to the Service Providers’ deployment. We consider traffic *ingressing* the deployment since, in the egress setting, the network has full control over traffic up until the egress point which is the part of the path the network wants to control. Existing systems direct egress traffic to optimize performance and mitigate congestion [12, 15, 5].

Even though Service Providers lack interdomain control, there is hope that they can handle dynamic conditions. Service Providers globally have enough capacity to handle even large changes in traffic, and can efficiently utilize this capacity during peak times since large loads are often localized, and since Service Providers deploy infrastructure and peer with networks in multiple locations around the world.

Figure 2 shows how a certain type of dynamic condition — failure — can lead to performance problems, even with *selectivecast* advertisements, and how Service Providers can avoid such overutilization with enough foresight. In nor-

mal operation, user traffic is split evenly across two prefixes. However when site B fails, all traffic ingresses through the same link since BGP chooses the route through Provider 1, causing overload (links have capacity 1). Advertising prefix 2 to provider 2 at site A *a priori* allows the Service Provider to split traffic between the two links during failure, avoiding overload.

2.5 Key Challenges

Since Service Providers connect globally with thousands of networks, there is sufficient available capacity across paths/sites to satisfy user demand even under dynamic conditions. Placing traffic demand on paths to optimize performance objectives subject to capacity constraints would therefore be simple *if we could expose all the paths*. However, exposing paths uses prefixes which are expensive [?], so we must find some subset of paths to expose.

Finding that right subset of paths to expose that satisfies performance objectives, however, is hard since interdomain routing is difficult to model and since traffic conditions/failures are hard to predict. Greedy approaches to solving the problem (*e.g.*, [?]) do not work because every iteration of greedy allocation is tightly coupled with all other iterations due to capacity constraints on links — a decision at one iteration can negatively interfere with decisions at all other iterations. Moreover, there are many failure conditions to consider, far too many to explicitly optimize for, and under each failure condition it is unclear how users will route to the deployment due to complex interdomain routing dynamics.

3 Methodology

3.1 Problem Setup and Definitions

3.1.1 High-Level Overview

We aim to advertise relatively few prefixes to connected networks to offer low-latency paths from user networks to the Service Provider subject to capacity constraints on links. We assess resilience based on single-link and single-site failures, but our methodology extends to the multi-link/site failure case and our evaluation shows that solutions that optimize for these metrics naturally offer other benefits (§5). Adding capacity constraints on sites (*e.g.*, servers) would be a straightforward extension.

We assume the Service Provider has some technology for directing traffic towards prefixes. Examples include DNS [3, 1?], multipath transport [? ?], or control-points at/near user networks [? ?]. DNS offers slow redirection due to caching [?], but is the most readily deployable by the largest number of networks, whereas service-provider-controlled appliances offer precise control but may not be a feasible option for some Service Providers. Service Providers with stronger incentives

to provide the best service to users will invest in better options with more control, and eventually multipath transport will see wide enough deployment to be used by all Service Providers—today, MPTCP is enabled by default in iOS [?] and Ubuntu 22 [?]. In our evaluations, we assume all users can be precisely directed to each prefix.

Service Providers connect to peers/providers at sites via physical connections we call peering links. Users route to the deployment through the public Internet to a prefix, over one of the peering links via which that prefix is advertised. The path (and therefore peering link) is chosen via BGP. We consider users at the granularity of user groups (UGs), which generally refer to user networks that route to the Service Provider similarly and experience similar latency, but could mean different things to different Service Providers (*e.g.*, /24 IPv4 prefixes, metros). UGs generate traffic volumes, v_{UG} , known to the Service Provider, and the Service Provider provisions capacity at links/sites to accommodate this load. We assume a system run by the Service Provider measures latency from users to prefixes advertised by the Service Provider (a reasonable assumption [2, 3, 4, 16?]). For a given user, paths towards prefixes may be different and so may have different latency.

3.1.2 Problem Formulation

As in prior work, we model an advertisement configuration A as a set of $\langle \text{peering}, \text{prefix} \rangle$ pairs where $\langle \text{peering}, \text{prefix} \rangle \in A$ means we advertise that prefix via that peering. We then seek an advertisement configuration that minimizes latency during normal operation (N) and single link/site failures (F_l) subject to capacity constraints, c_l . That is we aim to solve

$$\min_A \sum_{UG} L_{UG}(A, N) v_{UG} + \alpha \sum_l L_{UG}(A, F_l) v_{UG} \quad (1)$$

The latency a user group experiences under an advertisement configuration A under deployment N or F_l (representing the deployment with a component failure l) is the solution to the globally optimal allocation of user traffic to paths under A . A given advertisement configuration exposes a set of paths to UGs through peering links (l) which we uniquely identify with the (UG, l) tuple. This allocation is the solution to the following linear program for $w_{UG,l}$, the allocation of user traffic to paths.

$$\begin{aligned} \min_{w_{UG,l}} \quad & MLU(w_{UG,l}) + \beta \sum_{UG,l} L_{UG,l} w_{UG,l} \\ \text{s.t.} \quad & \sum_l w_{UG,l} = v_{UG} \quad \forall UG \\ & w_{UG,l} \geq 0 \end{aligned} \quad (2)$$

The minimization term is the sum of maximum link utilization (MLU) and latency, weighted by a parameter β .

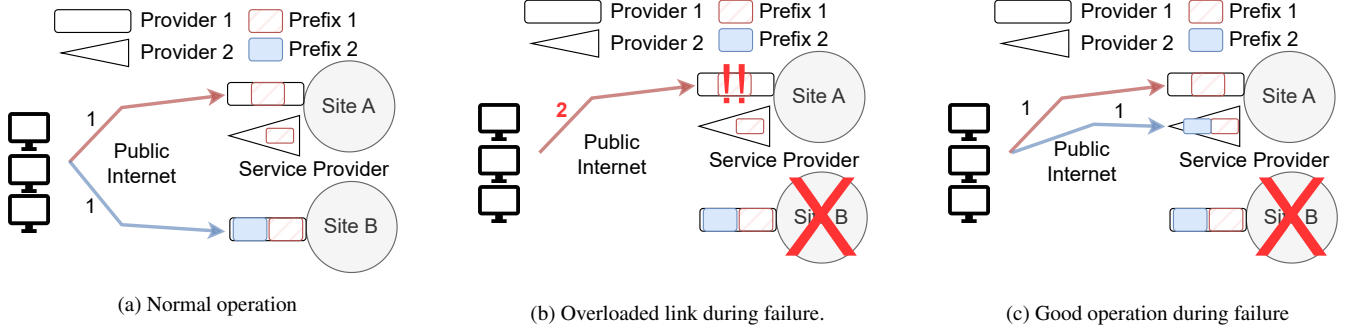


Figure 2: In normal operation traffic is split between two sites by directing half the traffic to each prefix (2a). When site B fails, there is enough global capacity to serve all traffic (each link can handle 1 unit) but no way to split traffic across multiple providers given available paths leading to link overload (2b). A *resilient* solution is to advertise prefix 2 to an additional provider at site A, allowing traffic splitting across the two links (2c).

represents a tradeoff between using uncongested links/sites and low-propagation delay, and is set by the Service Provider based on their goals. The constraints require that user volume on a link $w_{UG,l}$ be positive and sum to the users total volume v_{UG} across all links. Hence, optimizing Equation (2) amounts to finding low-latency assignments of UGs to links that do not cause congestion. By considering solutions to this problem in both normal operations, N , and under link/site failure, F_l , in Equation (1) we encourage low-latency, resilient advertisement strategies.

3.2 Approximations to Accommodate the Internets' Limitations

Solving Equation (1) is challenging because computing $L_{UG}(A, N)$ requires advertising prefixes in the Internet, which can only be done infrequently to avoid route-flap-dampening. Hence we model, instead of measure, user routes and improve this model over time through measurements.

3.2.1 Modeling User Routes

Representing both Equation (1) and Equation (2) as a single mixed-integer linear program (*i.e.*, solving directly for A) would be feasible if we knew exactly how users routed to all advertisements, but routing is hard to predict [9?].

We instead model routing probabilistically and update our probabilistic model over time as we measure how users route to the deployment, borrowing methods from prior work [16? ?]. That is, in Equation (1) we model $L_{UG}(A, N)$ probabilistically. Our probabilistic model assumes apriori, for a given UG towards a given prefix, that all ingress options for a user are equally likely. Upon learning that one ingress is preferred over the other, we exclude that less-preferred ingress as an option for that UG in all future calculations for all prefixes for which both ingresses are an option. As we exclude more options, latency/path distributions on unmeasured scenarios converge to the true latency/path. An example of this process is shown in Figure 3, where we refine our latency estimate towards an

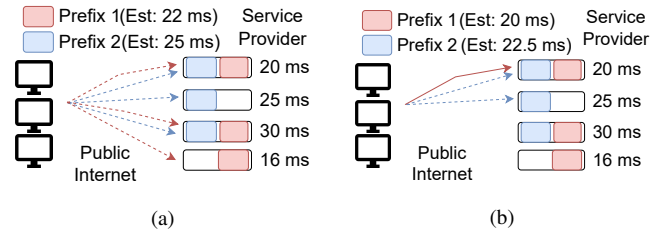


Figure 3: With no knowledge of routing preferences, we estimate latency from this user to both the red and blue prefixes with the average over possible ingress latencies (3a). We measure towards the red prefix and learn that the first ingress has higher preference than the third and fourth ingresses. We use this information to refine our latency estimate towards the blue prefix since the third ingress is no longer a possible option (3b), without measuring the blue path.

unmeasured prefix (blue) using measurements towards other prefixes (red).

3.2.2 Modeling User Decisions

To assign traffic to prefixes we solve Equation (2). However, during optimization we only know the distribution of possible paths on unmeasured advertisements so we cannot solve for optimal allocations of traffic to prefixes. Therefore, we instead approximate latency, $L_{UG}(A, N)$, on as-yet unmeasured advertisements A in Equation (1) by assigning user traffic to their lowest-latency paths. When we notice this lowest-latency assignment leads to overloaded links, we inflate $L_{UG}(A, N)$ proportionally to the oversubscription. Hence, in solving Equation (1) we implicitly penalize advertisement configurations that place too much traffic on any one link.

3.2.3 Modeling Latency Over Users

Since we model paths as probabilistic, how traffic we place traffic on paths, and therefore what latency users experience ($\sum_{UG} L_{UG}(A, N) v_{UG}$) is also probabilistic. We therefore model the objective function in Equation (1) as the *expected value* of the summations. That is, if the latencies for UG across differ-

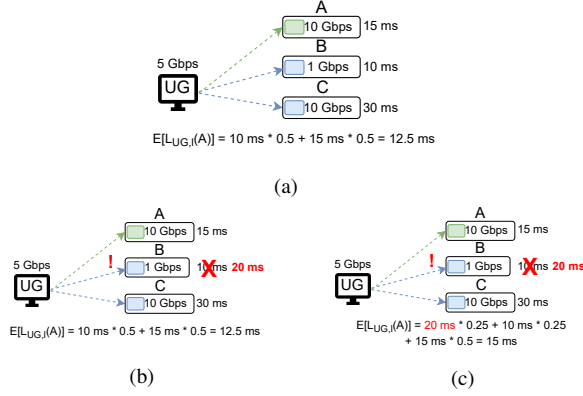


Figure 4: A UG has a path to two prefixes, green and blue. The green prefix is only advertised via ingress A and the blue prefix is advertised via two ingresses (B and C), each of which are equally likely for this UG. Hence the initial expected latency is the average of 10 ms and 15 ms corresponding to whether this UG prefers ingress B or C more (4a). However, ingress B does not have sufficient capacity to handle this UG’s traffic and so fails with 50% probability (4b). We artificially inflate this UG’s expected latency (and all other UG’s using ingress B) to reflect this possible overutilization (4c).

ent paths under advertisement A are $L_{UG,I}(A)$ we compute the distribution of the minimum choice, $\min_I L_{UG,I}(A)$, and then compute the distribution over the sum $\sum_{UG} \min_I L_{UG,I}(A) v_{UG}$. We use the minimum latency choice as a simple approximation of how traffic is directed to sites for which we can compute a probability distribution (as opposed to solving Equation (2), which we cannot do probabilistically).

It could be that such minimum-latency assignments lead to congested links, especially during link/site failures. We would like to penalize such advertisement scenarios, and favor those that distribute load more evenly without solving Equation (2). Hence, before computing the expected latency, we first compute the probability that links are congested by computing the distribution of link utilizations from the distribution of user assignments to paths. We then inflate latency for users on overutilized links.

That is, for each possible outcome of user assignments to links, we compute link utilizations and note the probability those users reach each link. We then accumulate the probability a link is congested as the total probability over all possible scenarios that lead to overutilization. We discourage users from choosing likely-congested links using a heuristic — we artificially inflate latency for users on those links proportional to the overutilization factor. After inflating user latencies, we recompute the distribution of average user latency (without changing user decisions), and take the expected value of this random variable in Equation (1) as a proxy for the true latency. Hence, this heuristic penalizes cases with lots of overutilized links. We show an example in Figure 4.

These heuristics tend to yield accurate estimates of overall benefit when averaged across the entire deployment (*i.e.*, Equation (2)), despite possible inaccuracies in predicting any

individual UG’s latency or link’s utilization.

3.3 Solving with Gradient Descent

We cannot solve Equation (1) directly/exhaustively since the optimization variable is an integer-matrix. Instead, we approximate the optimization variable, A , as a continuous variable with entries between 0 and 1 and threshold its entries at 0.5 to determine if a prefix is advertised/not advertised to a certain peer/provider. We then solve Equation (1) using gradient descent, approximating gradients between adjacent advertisements using sigmoids as in related work in the optimization literature **TBD: cite**.

Gradient descent is appropriate for two reasons: first, it is parallelizable which is important given the high dimensionality of A (tens to hundreds of thousands of entries representing all $\langle \text{peering}, \text{prefix} \rangle$ pairs). Second, it simultaneously weighs several (possibly opposing) goals in Equation (1) according to their global importance in minimizing the objective. Intuitively, Equation (1) has opposing goals since advertising prefixes to expose backup paths may push some users off their lowest-latency route, but may also encourage demand-spreading during failures which is more important than having “optimal” steady-state latency. Gradient descent takes into account both of these advantages across all $\langle \text{peering}, \text{prefix} \rangle$ pairs simultaneously.

We do not compute all entries of the gradient of Equation (1) since there are too many to compute. Instead, we use monte-carlo techniques to compute the expected gradient as in prior work **TBD: cite**.

Minimization scales linearly with the number of users, quadratically with the number of peers/providers, and linearly with the number of prefixes. Despite this high complexity, in practice the implementation runs quickly (minutes per gradient step) relative to the rate at which we can advertise prefixes (once an hour). Although solving Equation (1) does not have convergence guarantees, we have found that it finds good solutions over a wide range of simulated topologies, and converges quickly with thousands of UGs and $\langle \text{peering}, \text{prefix} \rangle$ pairs (§5).

3.4 Exploration to Improve Our Model

As, apriori, we do not know user preferences, we may be uncertain about the benefit of other advertisement strategies which could make solving our model using gradient descent noisy/unstable. For example, an adjacent strategy could have better *expected* latency than the current one but have much worse *actual* latency. Gradient descent would then push us towards a worse strategy.

To improve convergence, at each gradient step we compute benefit distributions on adjacent advertisement schemes and look for strategies for which we are very uncertain about whether it will be better or worse. One could use different

measures of uncertainty — we choose entropy. These measurements take time (as measuring all routes takes time), but improve convergence.

4 System Implementation

We prototype `SCULPTOR` on the `PEERING` testbed [?], which is now available at 32 Vultr cloud PoPs. We describe how we built `SCULPTOR` in the real Internet, and how we *emulate* a Service Provider including their clients, traffic volumes, and resource capacities. (We are not a Service Provider and so could not obtain actual volumes/capacities, but our extensive evaluations (§5) demonstrate `SCULPTOR`’s potential in an actual Service Provider, and our open/reproducible methodology provides value to the community.)

4.1 Simulating Clients and Traffic Volumes

To simulate client performances, we measured actual latency from IP addresses to our `PEERING` prototype as in prior work [?], and selected targets according to assumptions about Vultr cloud’s client base.

We first tabulate a list of 5M IPv4 targets that respond to ping in each /24 via exhaustive probing. We then find addresses that have routes through Vultr peers (as opposed to providers, for which every address has a route) by looking at Vultr’s routes. Vultr exports best routes from all PoPs to our `PEERING` client, from which we infer Vultr’s peers/providers from next-AS hops and BGP communities. Vultr documents how they tag different routes with different communities [?]. We then measure latency to each peer individually by advertising a prefix solely to that peer using Vultr’s BGP action communities, thus measuring client performance along many paths. We also measure performance from all clients to all providers, individually.

In our evaluations, we limit our focus to clients who had a route through at least one of Vultr’s peers. Vultr likely peers with networks with which it exchanges a significant amount of traffic so clients with routes through those peers are more likely to be “important” to Vultr. We found 600k /24s with routes through Vultr peers. We then select a subset of clients to limit the maximum number of clients with a route through each peer to limit computational complexity.

We then simulate traffic volumes according to anycast client PoP catchments, as in prior work. We first randomly choose the total volume of a PoP, and then divide that volume up randomly among clients whose anycast catchment was that PoP.

4.2 Simulating Resource Capacities

We assume that resource capacities are overprovisioned proportional to their usual load. However, we do not know the usual load of links and cannot even determine which peering

link traffic to one of our prefixes arrives on, as Vultr does not give us this information. (This limitation only exists since we are not a Service Provider, as a Service Provider could measure this using IPFIX, for example.) We overcome this limitation using two methods, each with their pros and cons.

First, we advertise prefixes into the Internet using the `PEERING` testbed, and measure actual ingress links to those prefixes using traceroutes from RIPE Atlas probes [13]. Specifically, we perform IP to AS mappings and identify the previous AS in the path to Vultr. This approach has limited evaluation coverage, as RIPE Atlas probes are only in a few thousand networks.

Second, we simulate user routes by assuming we know all users’ *preference models* (§3.2). We use a preference model where clients prefer peers over providers, and clients have a preferred provider. When choosing among multiple ingresses for the same peer/provider, clients prefer the lowest-latency option. We also add in random violations of the model. This second approach allows us to evaluate our model on all client networks, but may not represent actual routing conditions. However, we found that our key evaluation results held regardless of how we simulated routing conditions, suggesting our methodology is robust to such assumptions. Prior work also found the preference model to be valid in 90% of cases they studied.

Given either method we then measure anycast catchment and assign resource capacities as some overprovisioned percentage of this catchment. (Discussions with operators from CDNs suggested that they overprovision using this principle.) We choose an overprovisioning rate of 30%.

4.3 Deployments

We use a combination of real experiments and simulations to evaluate `SCULPTOR`. Both cases use simulated client traffic volumes, but our real experiments measure real routes using RIPE Atlas probes, while our simulations use simulated routes.

4.3.1 Experiments in the Internet

We assess how `SCULPTOR` performs in the real Internet using RIPE Atlas probes [13], which represent a subset of all clients. RIPE Atlas allows us to measure paths (and thus ingress links) to prefixes we announce from `PEERING`, which `SCULPTOR` needs to refine its model (??). However, RIPE Atlas does not have large coverage, as probes are only in 3,000 networks, and we are limited by RIPE Atlas probing rate constraints. Choosing RIPE Atlas probes to maximize coverage of networks and countries, we select probes from **TBD: X** networks which have paths to **TBD: Y** ingresses.

4.3.2 Simulations

We also assess how SCULPTOR by simulating user routes. Simulating user routes allows us to conduct more extensive evaluations as experiments take less time and consider more clients. We consider clients in **TBD: 10,000** networks to **TBD: 750** ingresses.

5 Evaluation

We thoroughly evaluate SCULPTOR on several dimensions and find it compares favorably to all other solutions during both steady-state and failure. Hence, SCULPTOR uses the same routing resources (prefixes, PoPs, links) more efficiently than other solutions, improving performance and reducing cost.

5.1 General Evaluation Setting

We compare SCULPTOR’s ability to give users low-latency routes to the Service Provider, subject to capacity constraints both during steady-state and failure conditions, to that of other solutions. The solutions include

anycast: A single prefix announcement to all peers/providers at all PoPs. Pure anycast is a very common strategy used in today’s deployments. **unicast**: A single prefix announcement to all peers/providers at each PoP. Another common strategy used in today’s deployments [7, 3]. **AnyOpt/ PAINTER**: Two strategies proposed in the literature for reducing steady-state latency compared to anycast. **One-per-Peering**: Advertise a unique prefix to each peer/provider, so all possible backup paths are always available to users. This solution provides a performance upper-bound, even though it is prohibitively expensive. (We obviously do not know the actual optimal solution with fewer prefixes.)

We use real latency measurements from **TBD: 70k** users to 750 peerings with simulated demands as described in ???. We then conduct evaluations both on the public Internet (*i.e.*, with real routing dynamics, ??) and in simulation (with simulated user ingress preferences, ??). Evaluating SCULPTOR in the public Internet demonstrates that SCULPTOR provides real, tangible benefits, but takes longer than simulation (as advertising prefixes is slow). Evaluating SCULPTOR using simulated routing dynamics allows us to conduct micro-evaluations in a variety of hypothetical scenarios.

For our evaluation in the public Internet, we **TBD: fill in details**.

For our simulated evaluations, we compute solutions over many random routing preferences, demands, and subsets of sites to demonstrate that SCULPTOR’s benefits are not limited to a specific deployment property. We evaluate SCULPTOR over deployments of size 3, 5, 10, 15, 20, 25, and 32 sites. Each size deployment was run at least **TBD: 10** times with random

user demands and routing preferences. We use twice the base-2 logarithm of the number of peers/providers as the number of prefixes for all solutions (except **One-per-Peering**). For deployments with more sites we use approximately 70 prefixes, and for smaller ones we use between 10 and 30. Prior work found that using this many prefixes to improve performance was a reasonable cost [16?].

In solving Equation (2), the lowest link utilization solution may have overloaded resources in failure scenarios. We assign all traffic arriving on the overloaded resource a very high latency but do not include this latency in quoted latency comparisons (actual latency would be a complicated function of congestion control protocols and queueing behavior). We comment separately on how much traffic is congested.

5.2 Simulation Results

In general, we find SCULPTOR to universally provide improvements over prior work in all studied metrics, but that these benefits often decrease as the number of sites increases.

5.2.1 SCULPTOR Finds Good Steady-State Paths

We first assess SCULPTOR’s ability to find low-latency paths for users in scenarios without failure. ?? demonstrates that SCULPTOR finds lower latency paths for more users than any other solution. The average latency improvement is **TBD: 6.4ms** compared to the next-best (PAINTER) of 9.6ms (45% improvement).

To generate ?? we solve Equation (2) for each solution and compare user latencies to the solution to Equation (2) using the **One-per-Peering** solution. Since resources have limited capacity, a user’s (globally) best ingress may be different from a user’s lowest-latency ingress. Anycast comparatively performs the worst with users being on average 25.3ms worse than **One-per-Peering** whereas SCULPTOR performs the best with users being 6.4ms worse than **One-per-Peering**. Interestingly, random (14.9ms worse) actually performs better than Anyopt (17.4ms worse) which could be due to the different settings (Anyopt was designed to optimize over a small number of provider connections).

5.2.2 SCULPTOR Improves Resilience to Failure

We next assess SCULPTOR’s ability to provide resilience to ingress-link failures and site failures which models any case where a link/site is no longer available; examples of such failures include excessive DDoS traffic on the link/site (thus using the link/site as a sink for the bogus traffic), physical failure, and/or planned maintenance. ?? demonstrates that SCULPTOR effectively shifts traffic without overloading alternate links/sites and without excessively penalizing performance.

To generate ?? we simulate link/site failures over a set of simulated deployments, each failure once, and compute UG

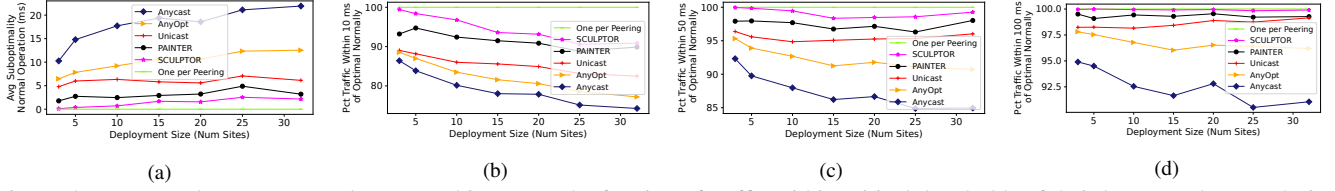


Figure 5: SCULPTOR lowers average latency and increases the fraction of traffic within critical thresholds of their best-case latency during normal operation.

to link allocations using Equation (2). For each failed component, we compute the difference between achieved and optimal (*i.e.*, One-per-Peering) latency for users who use that component in their steady-state solution. Hence each user is represented once in \mathbb{U} . We exclude failure scenarios for which even One-per-Peering experiences some overloading (20% of cases in single-PoP failure). Traffic landing on overloaded links is assigned a very large latency, so vertical intercepts on the left-hand side of the plot represent the fraction of congested users.

\mathbb{U} demonstrates that SCULPTOR finds backup lower-latency backup options than all other solutions (26ms worse than One-per-Peering on average). The next-best solution is unicast with 41.2ms worse on average, a 57% improvement. SCULPTOR further ensures that *no scenario* results in link overload, compared to unicast for which 5% of users experience congestion. anycast, AnyOpt, and random all overload a significant volume of traffic since none of these solutions guarantees backup options for users.

\mathbb{U} demonstrates that SCULPTOR similarly finds good backup options during PoP failure with latency 43.1ms worse than optimal, on average and congests 20% of users. Other solutions either offer higher-latency backup options (PAINTER: 49.6ms) or congest more users (unicast: 26). SCULPTOR is still far from optimal, and so improving it further is an area for future work. However, we note that it is a challenging problem given that One-per-Peering also cannot solve all scenarios without overloading.

It is interesting that unicast provides decent resilience despite its simplicity. Intuitively, this result is likely since unicast guarantees many backup paths for all users whereas PAINTER does not consider backup paths in its optimization. However, it pays no mind to whether these backup paths have available capacity and so does not perform as well as SCULPTOR.

5.2.3 SCULPTOR Improves Infrastructure Utilization

One response to increased link/site utilization is to install more capacity so that there is sufficient headroom to satisfy peak demand. We show that this response is not always necessary with better routing — SCULPTOR finds ways of distributing load over existing infrastructure to accommodate increased demand without adversely affecting latency.

In **TBD: fig**

5.3 Evaluating Convergence Properties

5.3.1 time until convergence

[tradeoff between time and benefit with number of iters and number of info searches]

5.4 Actual Deployment Results

6 Related Work

- Other/related solutions to the problem
 - Painter, tipsy, anyopt, fastroute, footprint, jiangchen, analyzing the performance of an anycast cdn
- Egress
 - Espresso, swan, edge-fabric
- Resilience
 - Hose-based cross-layer backbone network design with Benders decomposition
 - * Capacity-efficient and uncertainty-resilient backbone network planning with hose (obsolete tho)
 - EBB: Reliable and Evolvable Express Backbone Network in Meta
 - Evolve or Die: High-Availability Design Principles Drawn from Googles Network Infrastructure
 - Decentralized cloud wide-area network traffic engineering with {BLASTSHIELD}
- Congestion / problems on paths
 - Zooming in on Wide-area Latencies to a Global Cloud Provider
 - * Significant fractions of performance problems are in the “middle” of the path, and thus potentially fixable by switching to a different path
 - Anycast Agility: Network Playbooks to Fight DDoS
 - * Manually craft set of BGP actions to place certain volumes of traffic on certain sites
- 5g/multipath systems

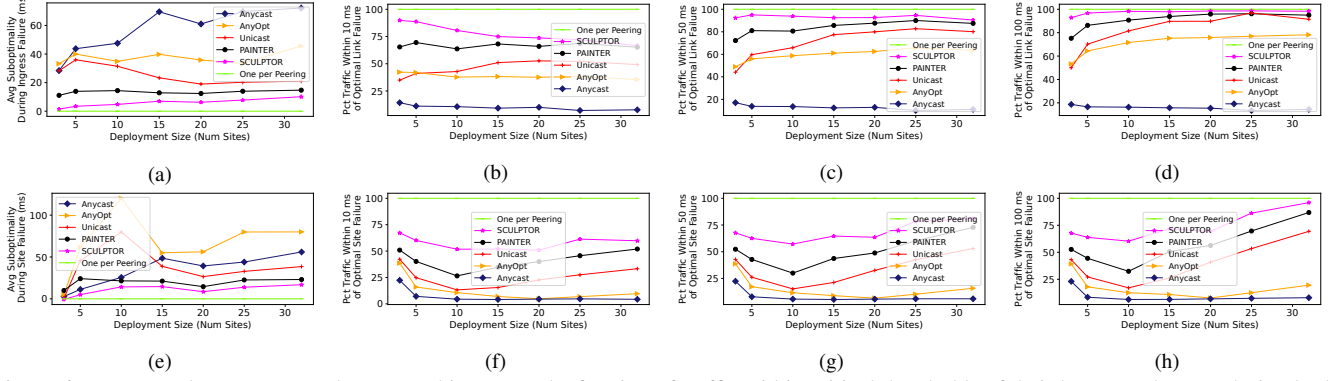


Figure 6: SCULPTOR lowers average latency and increases the fraction of traffic within critical thresholds of their best-case latency during both link and site failures.

- Evolving Mobile Cloud Gaming with 5G Standalone Network Telemetry
- Xlink

7 Conclusions

In conclusion

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