

Storj

A Decentralized Cloud Storage Network Framework

Alex Bender (bender@storj.io), Alex Leitner (alex@storj.io),
Benjamin Sirb (bens@storj.io), Braydon Fuller (braydon@storj.io),
Bryan White (bryan@storj.io), Chris Pollard (cpollard1001@gmail.com),
Dennis Coyle (dennis@storj.io), Dylan Lott (dylan@storj.io),
Garrett Ransom (garrett@storj.io), Gordon Hall (gordonhall@openmailbox.org),
James Hagans (jhagans@storj.io), James Prestwich (james@storj.io),
John Gleeson (jg@storj.io), Josh Brandoff (josh@brandoff.is), JT Olio (jt@storj.io),
Kaloyan Raev (kaloyan@storj.io), Kishore Aligeti (kishore@storj.io),
Nadine Farah (nadine@storj.io), Natalie Villasana (nat@storj.io),
Patrick Gerbes (patrick@storj.io), Philip Hutchins (philip@storj.io),
Shawn Wilkinson (shawn@storj.io), Tome Boshevski (tome@storj.io)

<https://github.com/storj/whitepaper>

June 24, 2018
v3.0

Abstract

Decentralized cloud storage is attractive for a number of reasons. Eliminating central control allows users to store and share data without reliance on a third-party storage provider. Decentralization mitigates many traditional data failures and outages while it simultaneously increases security and privacy. It allows market forces to innovate on cheaper ways to provide storage at a greater rate than any single entity can afford. While there are many ways to build such a system, there are some specific responsibilities any given system should address. Based on our experience with petabyte-scale storage systems, we introduce a modular framework for considering these responsibilities and building such a system, along with initial concrete implementations for each responsibility of the framework.

1 Introduction

Storj is a protocol that creates a distributed network for the reliable storage of data and facilitates payment for successful data storage between peers. The Storj protocol enables peers on the network to transfer data, verify the integrity and availability of remote data, retrieve data, and pay other nodes for storing data. Each peer is an autonomous agent, capable of performing these actions without significant human interaction.

Many storage products have been created based on distributed storage techniques. Products such as Wuala, Allmydata, Tahoe-LAFS, Space Monkey, Sia, Maidsafe, Filecoin, Crashplan, Mozy, HDFS, Storj, S3, GFS, all share one thing in common: a single computer is not as powerful or as robust as a network. These products and others attempt to solve many different use cases and have many different requirements, but at a high level, they all operate on the same principles. They generate redundancy for data in case of failure, store this redundancy in locations with varying degrees of failure isolation, then keep track of where the data was placed.

There are many different primary focuses one could choose when creating a distributed storage system. Speed, capacity, simplicity, trustlessness, byzantine fault tolerance, security, cost, etc., are all desirable traits in a storage system, but independently of anything else, data must be maintained to prevent data loss, nodes in the system must be able to be communicated with, and metadata must be kept track of.

While the Storj design space is different from systems such as HDFS or Wuala, we propose a framework that will allow us to choose some reasonable tradeoffs and then iterate on improvements to components of the system without changing the overall system. We accomplish this by breaking the design up into a collection of relatively independent concerns, and then bring them together to form the final protocol.

The rest of the paper is divided into 4 sections. In Section 2 we discuss the design space Storj operates in and we elaborate on some specific design constraints. Section 3 covers our framework and proposes a simple concrete implementation of each framework component. Section 4 covers specific details about how we will ship section 3 to users. Section 5 covers future areas of research.

2 Storj design constraints and considerations

Before designing a system, it's important to understand the requirements of said system, so we'll begin with a discussion of Storj's design constraints.

2.1 S3 compatibility

The flagship cloud storage product is Amazon's Simple Storage Service, or S3 for short. Most cloud storage products provide some form of compatibility with the S3 API.

Until a decentralized cloud storage protocol is the *lingua franca* of storage protocols, a graceful transition must be allowed for users with data currently on a centralized provider who are interested in the benefits of decentralized cloud storage but have low tolerance for switching costs.

For Storj to compete successfully in the wider cloud storage industry and bring decentralized cloud storage to the mainstream – thus enabling more people greater security and less centralized control – applications built against S3 should be able to be made to work with Storj with minimal friction and changes. This adds strong requirements on feature set, performance, and durability.

2.2 Byzantine Fault Tolerance

Unlike datacenter-based solutions like Amazon S3, Storj operates in an untrusted environment, where individual storage nodes are not necessarily assumed to be trustworthy. Storj operates over the public internet, and allows anyone to sign up to become a storage node.

We adopt the BAR (Byzantine, Altruistic, Rational) model [1] to discuss participants in the network. *Byzantine* nodes can depart arbitrarily from a proposed protocol whether it benefits them or not. *Altruistic* nodes participate in a proposed protocol even if the rational choice is to deviate. *Rational* nodes participate exactly when it is in their net benefit to do so, and may depart for the same reason.

Some distributed storage systems operate in an environment where all nodes are altruistic [add citation?](#). Storj operates in an environment where a majority of storage nodes are rational and a minority are byzantine. Any potential design must account for this distinction.

2.3 Device failure and churn

For any storage system, but especially a distributed storage system, component failure is a guarantee. All hard drives fail after enough wear [2], and the servers providing the network access to these hard drives will eventually fail, too. Network links die, power is lost, and storage mediums become unreliable. For data to outlast individual component failure, data must be stored with enough redundancy to recover from failure. Perhaps more importantly, no data should be assumed to be stationary and all data must eventually be moved. In such an environment, redundancy, data maintenance, repair, and replacement of lost redundancy are facts of life, and the system must account for these issues.

Decentralized systems are additionally susceptible to high churn rates, where potential participants join the network and then leave for various reasons, well before their hardware has actually failed. A network with a high churn rate will use a large amount of bandwidth just to ensure durability of the data, and such a network will fail to scale. As a result, a scalable, highly durable storage system must prefer stable nodes and endeavor to keep the churn rate as low as possible. Despite the issues with hardware failure and node churn, Maymounkov et al. found that in decentralized systems, the probability of a node staying on for an additional hour as a member of the network is an increasing function of uptime [3]. In other words, the longer a node is a participant in the network, the more likely it is to continue participating. See Appendix [TODO](#) for a discussion about how repair bandwidth varies as a function of node churn and uptime.

2.4 Latency

Decentralized, distributed storage has massive opportunities for parallelism with transfer rates, processing, and number of other factors. Parallelism by itself is a great way to increase overall throughput even when individual network links are slow. However, parallelism cannot by itself improve *latency*. If an individual network link has fixed latency and is a required part of an operation, the latency of the overall operation will be bound from below by the latency of the required network link. Therefore, a distributed system intended for high performance applications must aggressively optimize for low latency, both at the individual process scale and at the overall architecture scale.

We emphasize an architectural strategy aimed at achieving low latency by focusing on eliminating the need to wait for long tails [4]. The goal is a protocol that allows for every request to be satisfiable by the fastest nodes participating in any given transaction, without waiting for a slow subset. Focusing on operations

where the result is only dependent on the fastest nodes turns what could be a potential liability (highly variable performance from individual actors) into a great source of strength for a distributed storage network.

2.5 Bandwidth

Global bandwidth availability is increasing year over year, and access to high-bandwidth internet connections is unevenly distributed. While users in some countries can easily access symmetric, high-speed, unlimited bandwidth, users in other countries may have significant difficulty in obtaining access to the same. In the United States, many residential internet service providers provide internet in a way that presents two specific challenges for designers of a decentralized network protocol. The first challenge is that the internet connection is often asymmetric, where customers subscribe to internet service based on the advertised download speed, and the upload speed is potentially an order of magnitude or two slower. The second challenge is that bandwidth is sometimes "capped" at a fixed amount of traffic per month. Such caps impose significant limitations on the bandwidth available at any given moment. For example, an internet connection with a throughput of 10 MBPS with a cap of one terabyte per month may not average more than 380 KBPS (in a month) without going over the monthly bandwidth cap.

With device failure and churn guaranteed, any decentralized system will have a corresponding amount of repair traffic. It's therefore important to make sure there is enough headroom for the bandwidth required by data maintenance, over and above the bandwidth required for data storage and retrieval. Designing a storage system that is careless with bandwidth usage would be to relegate that system below storage providers with access to unlimited high-speed bandwidth, recentralizing the system to some degree. To keep the storage system as decentralized as possible and for it to work in as many environments as possible, bandwidth usage must be aggressively minimized.

Please see Appendix **TODO** for a discussion on how available bandwidth, combined with required repair traffic, limits usable space.

2.6 Security and privacy

TODO we want to make sure users' data privacy is protected

2.7 Object size

Broadly, large storage systems can be classified into two groups by average object size. When storing lots of small bits of information, generally a database is the preferred route. On the other hand, when storing lots of large files (where "large" files means a few megabytes or more), an object store or filesystem are ideal.

The initial product offering by Storj Labs is designed to function primarily as an object store for larger files. While future improvements may enable database-like use cases, the predominant use case described by this paper is object storage. Protocol design decisions are made with the assumption that the vast majority of objects stored will be a couple of megabytes or more. It is worth pointing out that this will not negatively impact use cases that require lots of files smaller than a megabyte, since such cases can admit a packing strategy, where many small files are aggregated and stored together as one large file. As the protocol has streaming support, small files can be retrieved without requiring full retrieval of any aggregated object they were packed into.

3 Framework and concrete implementation

At a high level, there are three major operations in the system: storing data, retrieving data, and maintaining data.

Storing data When data is stored with the network, the client encrypts it, breaks it up into multiple little pieces, distributes the pieces to peers in the network, and generates and stores some metadata about where to find the data again.

Retrieving data When data is retrieved from the network, the metadata about where to find the pieces is recovered first, then the pieces are retrieved and the original data is reconstructed on the client's local machine.

Maintaining data Data is maintained in the network with nodes replacing missing pieces when the amount of redundancy drops below a certain threshold. The data is reconstructed and the missing pieces are regenerated and replaced.

To make this system feasible while satisfying our design constraints, we will need to solve a number of complex challenges. Inspired by Raft [5], we break the

design up into a collection of relatively independent concerns and then combine them to form the desired protocol. One important benefit of doing so is this makes it much easier to update individual components without rearchitecting the rest of the network. The individual components are:

1. Storage nodes
2. Peer-to-peer communication
3. Overlay network
4. Redundancy
5. Structured file storage
6. Metadata
7. Encryption
8. Authorization
9. Data repair
10. Reputation
11. Payments

3.1 Storage nodes

The most basic building block is the storage node. The storage node stores and returns data provided to it. Nodes should provide reliable storage space, network bandwidth, and appropriate responsiveness. In return, nodes are rewarded for their participation. Storage nodes will be selected based on a number of different criteria. Ping time, latency, throughput, disk space, geographic location, legal restrictions, etc., are all important factors that may need to be considered. This means that node selection almost certainly must be an explicit process.

3.1.1 Concrete implementation

Storage nodes will support three methods: `get`, `put`, and `delete`. Storage nodes store *pieces* (to be described in more detail later). Each method will take a *piece ID*, a *payer ID* and signature by the payer, an optional TTL, and the other metadata required by the bandwidth accounting protocol (also to be described later).

The **put** operation will take a stream of bytes and store the bytes such that any subrange of bytes can be retrieved again via a **get** operation. **get** operations are expected to work until the TTL expires (if a TTL was provided), or until a **delete** operation is received, whichever comes first.

The *payer ID* forms a namespace. An identical *piece ID* with a different *payer ID* refers to a different *piece*.

Storage nodes should allow administrators to configure maximum allowed disk space usage and maximum allowed bandwidth usage over the last rolling 30 days. Storage nodes should keep track of how much is remaining of both. Storage nodes should reject operations that do not have a valid signature from the appropriate payer.

3.2 Peer-to-peer communication

All peers on the network will need to communicate. The framework requires a reliable and ubiquitous protocol that all peers speak that:

- provides peer reachability, even in the face of firewalls and NATs. This may require techniques like STUN, relays, etc.
- provides authentication, where each participant knows exactly the identity of the peer with whom they are speaking.
- provides privacy, where only the two peers know what transfers between them.

3.2.1 Concrete implementation

Initially, we'll be using gRPC [6] on top of TLS on top of μ TP [7] with added STUN functionality. Over time, we'll be replacing TLS to reduce round trips due to connection handshakes in situations where the data is already encrypted and forward secrecy isn't necessary. **TODO** See the Future Work section for more details.

As in S/Kademlia [8], the *node ID* will be the hash of a public key and will serve as a proof of work for joining the network. Unlike Bitcoin proof of work [9], the work will be dependent on how many *trailing* zero bits one can find in the hash output. This means that the node ID will still be usable in a balanced Kademlia [3] tree.

The node's master public/private key will only be used to operate a miniature certificate authority. The node's private key can thus remain in cold storage if needed. The node will sign keys and certificates to be used for peer-to-peer communication. When using TLS, every peer can ascertain the ID of the node it is speaking with by validating the certificate chain and hashing the certificate authority's key. The peer can estimate how much work went into constructing that node ID by considering the number of 0 bits at the end of the ID.

For the few cases where a node cannot achieve a successful hole punch through a NAT or firewall via STUN, uPnP, NATPmP, or a similar technique, manual intervention and port forwarding will be required.

3.3 Overlay network

If, given a peer's network address, any other peer can connect to it, the framework requires a system to look up peer network addresses by node ID in the first place. An *overlay network* can be built on top of our peer-to-peer communication component that provides functionality similar to DNS, where a node's ID can be resolved to an ephemeral network address for communication.

3.3.1 Concrete implementation

The Kademlia DHT serves as a key-value store with a built-in node lookup protocol. We utilize this protocol to achieve DNS-like functionality for node lookup, while ignoring the storage aspects of the Kademlia protocol due to some issues around value republishing, limits to network growth rate, and so on. However, using a DHT will make it difficult to achieve millisecond-level response times when multiple DHT lookups must happen for every operation, so more work is necessary to achieve our performance goals. Fortunately, caching address information for an entire network of 80k nodes (for example) can be done with 3MB of memory, so the DHT can be sped up with some simple, optional caching.

Because a cache of the DHT can be untrusted (and peer-to-peer communication is authenticated to avoid man-in-the-middle-attacks anyway), some well-known community-run DHT caches can be provided that simply attempt to talk to every storage node every so often, evicting nodes from their cache that have not been seen recently. Since nodes are expected to be long lived with good uptime, they are expected to have stable addresses that don't change often. Thus, such a cache will add a massive performance boost, even when slightly

stale. In addition, the protocol will be resilient against an expected degree of node churn, so having a small number of stale addresses in a DHT cache will not alter the expected performance of the network. Furthermore, we avoid a number of known attacks by using the S/Kademlia extensions.

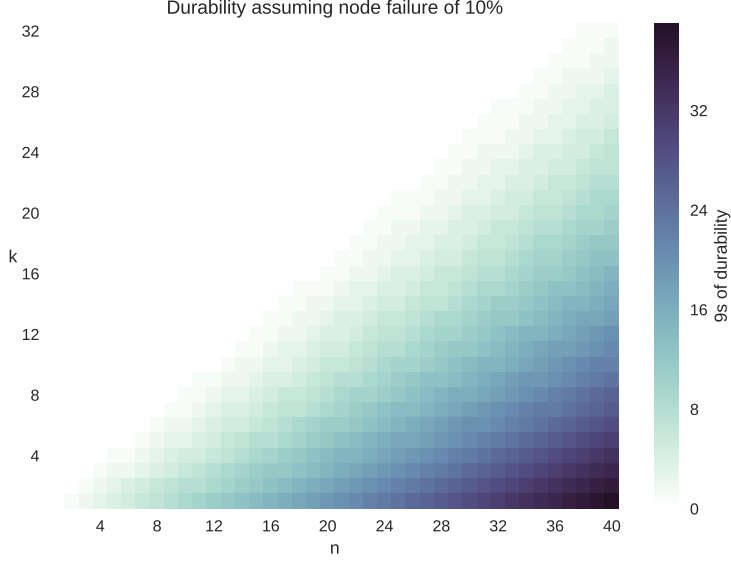
3.4 Redundancy

At any moment, any storage node could go offline permanently. Our redundancy strategy must store data in a way that provides access to the data with high probability, even though any given number of individual nodes may be offline. To achieve a certain level of *durability* (the probability that data will remain available in the face of failures), many products in this space use simple replication. Unfortunately, this ties durability to the network *expansion factor*, which is the storage overhead for reliably storing data.

As an example, suppose a certain desired level of durability requires a replication strategy that makes eight copies of the data. This yields an expansion factor of 8x, or 800%. This data then needs to be stored on the network, using bandwidth in the process. Thus, more replication results in more bandwidth usage for a fixed amount of data. As discussed in the protocol design constraints, high bandwidth usage prevents scaling, so this is an undesirable strategy for ensuring a high degree of file durability. Instead, *erasure codes* are a more general and more flexible scheme for manipulating data durability without tying it to bandwidth usage. Importantly, erasure codes allow changes in durability without changes in expansion factor!

An erasure code is often described by two numbers, k and n . If a block of data is encoded with a (k, n) erasure code, there are n total generated *erasure shares*, where only any k of them are required to recover the original block of data. If a block of data is s bytes, each of the n erasure shares is roughly s/k bytes. Besides the case when $k = 1$ (replication), all erasure shares are unique. Interestingly, the durability of a $(k = 20, n = 40)$ erasure code is better than a $(k = 10, n = 20)$ erasure code, even though the expansion factor (2x) is the same for both! Intuitively, this is because the risk is spread across more nodes in the $(k = 20, n = 40)$ case. These considerations make erasure codes an important part of our general framework.

With the simplifying assumption that every node has an equal probability of failure p , one can gain more intuition by modeling file durability as the CDF of the binomial distribution, as seen in Equation (1). In this case, $P(D)$ represents the probability that at most $n - k$ erasure shares are lost for a given file, so that



$n - (n - k) = k$ erasure shares of the file remain; i.e. the file can still be rebuilt. The CDF of the binomial distribution is given by:

$$P(D) = \sum_{i=0}^{n-k} \binom{n}{i} p^i (1-p)^{n-i} \quad (1)$$

By being able to tweak the durability independently of the expansion factor, very high durabilities can be achieved with surprisingly low expansion factors. Because of how limited bandwidth is as a resource, eliminating replication as a strategy entirely and using erasure codes only for redundancy causes a drastic decrease in bandwidth footprint and a drastic increase in the funds available per byte on storage nodes.

k	n	Exp. factor	P(D)
2	4	2	99.62999999999996341%
4	8	2	99.956834999999999436%
8	16	2	99.999407567699549748%
16	32	2	99.99999871786737771%
32	64	2	99.999999999999988898%

Table 1: $P(D)$ for various choices of k and n

3.4.1 Streaming

Erasures codes are used in many streaming contexts such as audio CDs and satellite communications, so it's important to point out that using erasure coding in general does not make our streaming design requirement more challenging. Whatever erasure code is chosen for our framework, streaming can be added on top by encoding small portions at a time, instead of attempting to encode a file all at once. See the structured file storage section for more details.

3.4.2 Long tails

Erasures codes enable an enormous performance benefit, which is the ability to avoid waiting for long-tail response times [4]. For uploads, a file can be encoded to a higher (k, n) ratio than necessary for durability guarantees. During an upload, after enough pieces have uploaded to gain required redundancy, the remaining additional uploads can be canceled, allowing the upload to be blocked by the fastest nodes in a set, instead of waiting for the slowest nodes. Downloads are similarly improved. Since more redundancy exists than is needed, downloads can be served from the fastest peers, eliminating a wait for temporarily slow or offline peers.

3.4.3 Concrete implementation

We use the Reed-Solomon erasure code. See [TODO Appendix](#) for how we select our Reed-Solomon numbers.

3.5 Structured file storage

Our design constraints include S3 compatibility. This means we should support hierarchical objects (paths with prefixes), object metadata, arbitrarily large files, arbitrarily large amounts of files, and so on. Similarly, our design constraints require security, so any such metadata must be encrypted.

Provided we have an efficient way to store data, we can build many of these features on top by means of an *injective embedding* (here used in the mathematical sense). Because so many details here depend on concrete implementation details, our framework is loose in specificity, while our concrete implementation has significant detail.

3.5.1 Concrete implementation

Bucket A **bucket** is an unbounded but named collection of **files** identified by **paths**. Each **path** represents one **file**, and every **file** has a unique **path**.

Path A **path** is a unique identifier for a **file** within a **bucket**. A **path** is a string of UTF8 codepoints that begins with a forward slash and ends with something besides a forward slash. More than one forward slash (referred to as the **path separator**) separate **path components**.

An example path might be `/etc/hosts`, where the **path components** are `etc` and `hosts`.

Clients encrypt **paths** before they ever leave the client computer.

File A **file** is a collection of **streams**. Every **file** has exactly one default **stream** and may have 0 or more named **streams**. Multiple **streams** allow flexible support of extended attributes, alternate data streams, resource forks, and other slightly more esoteric filesystem features.

Like **paths**, the data contained in a **file** is encrypted before it ever leaves the client computer.

Stream A **stream** is an ordered collection of 0 or more **segments**. **segments** have a fixed maximum size, and so the more bytes the **stream** represents through **segments**, the more **segments** there are.

Segment A **segment** represents a single array of bytes, between 0 and a user-configurable maximum **segment** size. Breaking large **files** into multiple **segments** provides a number of security and scalability advantages.

Inline Segment An **inline segment** is a **segment** that is small enough it makes sense to store it "inline" with the metadata that keeps track of it, such as a **pointer**.

Remote Segment A **remote segment** is a larger **segment** that will be encoded and distributed across the network. A **remote segment** is larger than the metadata required to keep track of its book keeping.

Stripe A **stripe** is a further subdivision of a **segment**. A **stripe** is a fixed amount of bytes that is used as an encryption and erasure encoding boundary size. Encryption and erasure encoding happen on **stripes** individually. Encryption happens on all **segments**, but erasure encoding only happens on **remote segments**.

Erasure Share When a **segment** is a **remote segment**, its **stripes** will get erasure encoded. When a **stripe** is erasure encoded, it generates multiple

pieces called **erasure shares**. Only a subset of the **erasure shares** are needed to recover the original **stripe**, but each **erasure share** has an index identifying which **erasure share** it is (e.g., the first, the second, etc.).

Piece When a **remote segment's stripes** are erasure encoded into **erasure shares**, the **erasure shares** for that **remote segment** with the same index are concatenated together, and that concatenated group of **erasure shares** is called a **piece**. If there are n **erasure shares** after erasure encoding a **stripe**, there are n **pieces** after processing a **remote segment**. The i th **piece** is the concatenation of all of the i th **erasure shares** from that **segment's stripes**.

Piece Storage Node A node in the network that is responsible for storing **pieces**. These are operated by **farmers**.

Farmer A person or group that is responsible for running and maintaining **piece storage nodes**.

Pointer A **pointer** is a data structure that keeps track of which **piece storage nodes** a **remote segment** was stored on, or the **inline segment** data directly if applicable.

3.5.2 Files as Streams

Many applications benefit from being able to keep metadata alongside files. For example, NTFS supports "alternate data streams" for each file, HFS supports resource forks, EXT4 supports "extended attributes," and more importantly for our purposes, AWS S3 supports "object metadata" [10]. Being able to support arbitrarily named sets of keys/values dramatically improves compatibility with other storage platforms. Every **file** will have at least one **stream** (the default **stream**) and many files may never have another **stream**.

3.5.3 Streams as Segments

Because **streams** are used for data (the default **stream**) and metadata (extended attributes, etc.), **streams** should be designed both for small data and large data. If a **stream** only has very little data, it will have one small **segment**. If that **segment** is smaller than the metadata it would require to be stored on the network, the **segment** will be an **inline segment** and the data will be stored directly inline with the metadata.

For larger **streams** past a certain size, the data will be broken into multiple large **remote segments**. Segmenting in this manner has a number of advantages to security, privacy, performance, and availability.

Maximum **segment** size is a configurable parameter. To preserve privacy, it is recommended that **segment** sizes be standardized as a byte multiple, such as 8 or 32 MB. Smaller **segments** may be padded with zeroes or random data. Standardized sizes help frustrate attempts to determine the content of a given **segment** and can help obscure the flow of data through the network.

Segmenting large files like video content and distributing the **segments** across the network separately reduces the impact of content delivery on any given node. Bandwidth demands are distributed more evenly across the network. In addition, the end-user can take advantage of parallel transfer, similar to BitTorrent [11] or other peer-to-peer networks.

3.5.4 Segments as Stripes

In many situations it's important to be able to access just a portion of some data. Some large file formats such as large video files, disk images, or file archives support the concept of seeking, where only a partial subset of the data is needed for correct operation. In these cases it's useful to be able to decode and decrypt only parts of a file.

A **stripe** is no more than a couple of kilobytes, and encrypting and encoding a single **stripe** at a time allows us to read portions of a large **segment** without retrieving the entire **segment**, allows us to stream data into the network without staging it beforehand, and enables a number of other useful features. **Stripes** should be encrypted client-side before being erasure encoded. The reference implementation uses AES256-GCM by default, but XSalsa20+Poly1305 is also provided. This protects the content of the data from the **farmer** housing the data. The data owner retains complete control over the encryption key, and thus over access to the data.

It's important to use authenticated encryption to defend against data corruption (willful or negligent), and with a monotonically increasing nonce to defeat reordering attacks. The nonce should be monotonically increasing across **segments** throughout the **stream**. If **stripe** i is encrypted with nonce j , **stripe** $i + 1$ should be encrypted with nonce $j + 1$. Each **segment** should get a new encryption key whenever the content in the **segment** changes to avoid nonce reuse.

3.5.5 Stripes as Erasure Shares

Erasure encoding gives us the chance to control network durability in the face of unreliable **piece storage nodes**. Erasure encoding schemes often are described as (k, n) schemes, where k **erasure shares** are needed for reconstruction out of n total. For every **stripe**, n **erasure shares** are generated, where the network has an expansion factor of $\frac{n}{k}$.

For example, let's say a **stripe** is broken into 40 **erasure shares** ($n = 40$), where any 20 ($k = 20$) are needed to reconstruct the **stripe**. Each of the 40 **erasure shares** will be $\frac{1}{20}$ th the size of the original **stripe**. All n **erasure shares** have a well defined index associated with them. The i th share will always be the same, given the same input parameters.

Because peers generally rely on separate hardware and infrastructure, data failure is not correlated. This implies that erasure codes are an extremely effective method of securing availability. Availability is proportional to the number of nodes storing the data.

See section **TODO** for a breakdown of how varying the erasure code parameters affects availability and redundancy.

3.5.6 Erasure Shares as Pieces

Because **stripes** are already small, **erasure shares** are often much smaller, and the metadata to keep track of all of them separately would be immense relative to their size. Instead of keeping track of all of the shares separately, we pack all of the **erasure shares** together into a few **pieces**. In a (k, n) scheme, there are n **pieces**, where each **piece** i is the ordered concatenation of all of the **erasure shares** with index i . As a result, where each **erasure share** is $\frac{1}{k}$ th of a **stripe**, each **piece** is $\frac{1}{k}$ th of a **segment**, and only k **pieces** are needed to recover the full **segment**.

3.5.7 Pointers

TODO

3.6 Metadata

In the previous section, we discussed how we will break up files, encode them for redundancy, and then store them in the network. Independently of the concrete organization and structure of this scheme, there are two types of metadata that are important to store somewhere for recovery: paths and what storage nodes received pieces (pointers).

Our framework requires a relatively performant system that can store pointers by path in a way that supports ordered iteration over those paths. Every time an object is added, edited, or removed, one or more entries in this metadata storage system will need to be adjusted. As a result, there could be heavy churn in this metadata system, and across the entire userbase the metadata itself could end up being a sizeable amount of data.

To talk more about the scope and scale we expect with some examples, suppose in a few years this system stores 1 total exabyte of data, where the average object size is 50MB and our erasure code is such that $n = 40$. Each object will use just one segment, and thus have one pointer each. The pointer will contain information about the segment encoding, including what n nodes the segment pieces are stored on. 1 exabyte of 50MB objects is 20 billion objects. If each pointer is roughly $40 \cdot 64 + 192$ bytes (info for each node plus the path and some general overhead), there are over 55 terabytes of metadata to keep track of (which is still 18,181 times less data to keep track of than an exabyte). Fortunately, this metadata can be heavily partitioned by user. A user storing a 100 terabytes of 50MB objects will only incur an overhead of 5.5 gigabytes, once again 18,181 times less data. It's worth pointing out that these numbers vary heavily with average object size (the larger the object size, the less the metadata overhead).

One of our framework's primary focuses is making sure this component – metadata storage – is interchangeable per user. Specifically, we expect to ship with multiple implementations of metadata storage that we will allow users to choose between. Other systems have spent an enormous amount of time attempting to solve this problem. We've concluded that multiple *good enough* solutions already exist, and propose using them.

Aside from scale requirements, the desired API is straightforward and simple: Put (store a pointer given a path), Get (retrieve a pointer given a path), List (paginated, ordered listing of existing paths), and Delete (remove a path).

3.6.1 Aside about distributed consensus

A long and challenging area of research has been directed toward getting a group of computers to agree on a set of values, with the goal of constructing a horizontally-scalable database that works in the face of expected failures (crash failures, for example: failures where a server simply shuts down). Fortunately, this research has led to some really exciting technology.

The biggest issue with getting a group of computers to agree is that messages can be lost. How this impacts decision making is succinctly described by the "Two Generals' Problem" [12] (earlier described as a problem between groups of gangsters [13]), in which two armies try to communicate in the face of potentially lost messages. Both armies have already agreed to attack a shared enemy, but have yet to decide on a time. Both armies must attack at the same time or else failure is assured. Both armies can send messengers, but the messengers are often captured by the enemy. Both armies must know what time to attack and that the other army has also agreed to this time.

Ultimately, a solution to the two generals' problem with a finite number of messages has been proven to be impossible, so engineering approaches have had to brace uncertainty by necessity. Many distributed systems make trade-offs to deal with this uncertainty. Some systems embrace *consistency*, which means that the system will choose downtime over inconsistent answers. Other systems embrace *availability*, which means that the system chooses potentially inconsistent answers over downtime. The widely-cited CAP theorem [14] states that every system must choose only two of consistency, availability, and partition tolerance. Due to the inevitability of network failures, partition tolerance is non-negotiable, and when a partition happens, every system must choose to sacrifice either consistency or availability. Many systems sacrifice both (sometimes by accident).

In the CAP theorem, consistency means that every read receives the most recent write or an error, so an inconsistent answer means the system returned something besides the most recent write without obviously failing. More generally, there are a number of *consistency models* that may be acceptable by making various tradeoffs. Linearizability, sequential consistency, causal consistency, PRAM consistency, eventual consistency, read-after-write consistency, etc., are all models for discussing how a history of events appears to various participants in a distributed system.¹

¹If differing consistency models are new to you, it may be worth reading about them in Kyle Kingbury's excellent tutorial [15]. If you're wondering why computers can't just use the current time to order events, keep in mind it is exceedingly difficult to get computers to even agree on that [16].

Amazon S3 generally provides *read-after-write consistency*, though in some cases will provide *eventual consistency* instead [17]. Arguably, there may be some flexibility here for the selection of alternate consistency models that suit us better while still broadly providing S3 compatibility. Many distributed databases provide eventual consistency by default, such as Dynamo [18] and Cassandra [19].

Linearizability in a distributed system is often much more desirable, as it is useful as a building block for many higher level data structures and operations such as distributed locks and other coordination techniques. Initially, early efforts centered around two-phase commit, then three-phase commit, which both suffered due to issues similar to the two generals' problem. Things were looking bad in 1985 when the FLP-impossibility paper [20] proved that no algorithm could reach linearizable consensus in bounded time. Then in 1988, Barbara Liskov and Brian Oki published the Viewstamped Replication algorithm [21] which was the first linearizable distributed consensus algorithm. Unaware of the VR publication, Leslie Lamport set out to prove linearizable distributed consensus was impossible [22], but instead in 1989 proved it was possible by publishing his own Paxos algorithm [23], which for some reason became significantly more popular. Ultimately both algorithms have a large amount in common.

Despite Lamport's claims that Paxos is actually simple [24], many papers have been published since then challenging that assertion. Google's description of their attempts to implement Paxos landed in Paxos Made Live [25], and Paxos Made Moderately Complex [26] is an attempt to try and fill in all the details of the protocol. The entire basis of the Raft algorithm is rooted in trying to wrangle and simplify the complexity of Paxos [5]. Ultimately, after an upsetting few decades, reliable implementations of Paxos, Raft, Viewstamped Replication [27], Chain Replication [28], and Zab [29] now exist, with ongoing work to improve the situation further [30, 31]. Arguably, part of Google's early success was in spending the time to build their internal Paxos-as-a-service distributed lock system, Chubby [32]. Most of Google's most famous internal data storage tools such as Bigtable [33] depend on Chubby for correctness. Spanner [34] – perhaps one of the most incredible distributed databases in the world – is mainly just two-phase commit on top of multiple Paxos groups.

Reliable distributed consensus algorithms have been game-changing for many applications needing fault-tolerant storage.

3.6.2 Aside about Byzantine distributed consensus

As mentioned in our design constraints, we expect most nodes to be *rational* and some to be *byzantine*, but few-to-none to be *altruistic*. Unfortunately, all of the previous algorithms we discussed assume a collection of altruistic nodes.

There have been a number of attempts to solve the Byzantine fault tolerant distributed consensus problem [9, 35–51]. Each of these algorithms make some additional tradeoffs the non-Byzantine distributed consensus algorithms don't require to deal with the potential for uncooperative nodes. For example, PBFT [35] causes a significant amount of network overhead. Bitcoin [9] intentionally limits the transaction rate with changing proof-of-work difficulty, in addition to requiring all participants to keep a full copy of all change histories (like other blockchain-based solutions).

TODO talk about merkle-dag, git-inspired approaches to metadata, potentially built on kademia, potentially reworking this entire section because ugh

3.6.3 Concrete implementation

Given the situation described in the asides about distributed consensus, we have decided that good-enough solutions already exist, so we will revisit the problem of solving Byzantine distributed consensus for our use case to a later release. We believe that a great distributed algorithm here is possible, all of the necessary building blocks are likely described above, and we expect to invest heavily in research to find it after we have a thriving user base with our solution based on good-enough approaches.

The most trivial implementation for the metadata storage functionality we require would be to simply have each user use their preferred trusted database such as PostgreSQL, SQLite, MongoDB, Cassandra [19], Spanner [34], CockroachDB, or something else. In many cases, this will be acceptable for specific users, provided those users were managing appropriate backups of their metadata. Indeed, the types of users who have petabytes of data to store probably can manage reliable backups of a single relational database storing only metadata.

There are a few downsides with this punt-to-the-user approach, however, such as:

- **Availability** - the availability of the user's data is tied entirely to the availability of their metadata server. The counterpoint here is that the

availability can be made arbitrarily good with existing trusted distributed solutions such as Cassandra, Spanner, or CockroachDB. Further, any individual metadata service downtime does not affect the entire network. In fact, the network as a whole can still never go down.

- **Durability** - if the metadata server suffers a catastrophic failure without backups, all of the user's data is gone. This is already true with encryption keys, but a punt-to-the-user solution increases the risk area from just encryption keys considerably. Fortunately, the metadata itself can be periodically backed up into the Storj system, such that only needing to keep track of metadata-metadata further decreases the amount of critical information that must be stored elsewhere.
- **Trust** - the user has to trust the metadata server.

On the other hand, there are a few upsides:

- **Use cases** - in a catastrophic scenario, this design still covers all required use cases.
- **Control** - the user is in complete control of all of their data. There is still no organizational single point of failure. The user is free to choose whatever metadata store with whatever tradeoffs they like. Like Mastodon [52], this solution is still decentralized. Further, in a catastrophic scenario, this design is no worse than most other technologies or techniques application developers frequently use (databases).
- **Simplicity** - other projects have spent multiple years on shaky implementations. We can get a useful product to market without doing this work at all. This is a considerable advantage.

Our launch goal is to allow customers to store their metadata in a database of their choosing. We expect and look forward to new systems and improvements specifically in this component of our framework.

3.7 Encryption

3.7.1 Concrete implementation

3.8 Authorization

3.8.1 Concrete implementation

3.9 Data repair

TODO thresholds **TODO** To prevent loss of the file, data owners should set shard loss tolerance levels. Consider a 20-of-40 erasure coding scheme. A data owner might tolerate the loss of 5 shards out of 40, knowing that the chance of 16 more becoming inaccessible in the near future is low. However, at some point the probabilistic availability will fall below safety thresholds. At that point the data owner must initiate a retrieve and rebuild process.

Because node uptimes are known via the audit process, tolerance levels may be optimized based on the characteristics of the nodes involved. Many strategies may be implemented to handle this process.

3.9.1 Concrete implementation

3.10 Reputation

3.10.1 Concrete implementation

3.11 Payments

3.11.1 Concrete implementation

3.12 Encryption

TODO

3.13 Structured file storage

3.14 Network state overview

The network state is a component of the distributed network, and has a primary role of storing information related to the segment. The network state keeps track of **pointers**, which contain the IDs of **piece storage nodes** holding **remote segments**, indication if the segment is **inline** and other descriptors. The network state service houses a **pointer** database that can be called by other services to put, get, list, and delete **pointers**.

Two components of the distributed network depend on the network state: the light client, and repair and maintenance (also known as data repair). The light client makes requests to the network state to store or receive **pointers** to **segments** depending on whether there is an upload or download action. Similarly, the data repair will request and receive a list of **pointers**. Each **pointer** contains the metadata mentioned above, including the following values related to Reed Solomon encoding: the minimum number of **pieces** required for a **segment**'s reconstruction, the total amount of **pieces** generated through erasure encoding, the repair threshold (or number of pieces needed to lose before triggering repair), and the success threshold (or amount of pieces needed to be stored in order to be counted as a success). Below are descriptions of how the light client and data repair interact with the network state for uploads, downloads, and data repair, respectively.

3.14.1 Download overview

In order to download a file, the light client will login and send a request to the network state to get pointers to the stored segments and other metadata. The light client extracts the node IDs that are storing the data and sends a request to the overlay network with those IDs. The overlay network responds with an object containing the nodes' IP addresses. Finally, the light client sends a request to farmers in order to receive pieces using those IP addresses.

Technical Dive into how Download Works **TODO**

3.14.2 Upload overview

In order to upload a file, the light client will log in and ask the overlay network for a set of farmers that fulfill a criteria, i.e. storage availability and bandwidth

to potentially store files. The overlay network will have a list of nodes that meet those criteria and use the reputation network to filter out reputable farmers. The overlay network then sends a refined list of farmers to the light client, which is then used to directly upload data. Finally, the light client sends a request to the network state in order to store the IDs of the piece storage nodes holding the segments and other related metadata in pointers.

Technical Dive into how Upload Works **TODO**

3.14.3 Repair and maintenance

The repair and maintenance component (also known as data repair) is essential to ensure data integrity is maintained. It confirms that nodes responsible for pieces continue to store the data that the light client sent. To do this, it first makes a request to the network state to receive pointers. From there, the repair and maintenance component extracts the node IDs from the response and makes a request with that information to the DHT cache. The DHT cache sends a response containing only the online nodes from the original node ID list. The data repair component takes the response from the DHT cache and the repair threshold value from the pointer, then calculates which pieces need to get repaired. The light client directly downloads pieces that need to get repaired from farmers and re-uploads them to new reputable farmers. To ensure the network state has up-to-date information about the data, the light client then sends a put request of new pointers with the updated node IDs.

Diagram Technical Detail *[WIP]*

3.15 Authorization

TODO

3.16 Farmer reputation

TODO

3.17 Payments

TODO

3.18 Payer reputation

TODO

3.19 Repair and maintenance

TODO discussion about audits using erasure codes instead of challenge merkle trees
TODO discussion about spot checks

TODO

Partial audits provide only probabilistic assurance that the farmer retains the entire file. They allow for false positive results, where the verifier believes the farmer retains the intact shard, when it has actually been modified or partially deleted. The probability of a false positive on an individual partial audit is easily calculable (see Section 6.4)

Thus the data owner can have a known confidence level that a shard is still intact and available. In practice, this is more complex, as farmers may implement intelligent strategies to attempt to defeat partial audits. Fortunately, this is a bounded problem in the case of iterative audits. The probability of several consecutive false positives becomes very low, even when small portions of the file have been deleted.

3.20 Payment

TODO Storj is payment agnostic. Neither the protocol nor the contract requires a specific payment system. The current implementation assumes Storjcoin, but many other payment types could be implemented, including BTC, Ether, ACH transfer, or physical transfer of live goats.

4 Product details

TODO The client provides the actual storage API. It should be run co-located with wherever data is generated, and will communicate directly with storage nodes so as to avoid central bandwidth costs. The client can provide an S3-compatible API to interoperate with existing software.

When data is sent to the client, the client will encrypt the data, choose

appropriate storage nodes, store the data on those nodes, then use the metadata system to keep track of which nodes the data resides on for future reading.

4.1 Farmer

TODO

4.2 Bridge

TODO As should be apparent, the data owner has to shoulder significant burdens to maintain availability and integrity of data on the Storj network. Because nodes cannot be trusted, and hidden information like challenge sets cannot be safely outsourced to an untrusted peer, data owners are responsible for negotiating contracts, pre-processing shards, issuing and verifying audits, providing payments, managing file state via the collection of shards, managing file encryption keys, etc. Many of these functions require high uptime and significant infrastructure, especially for an active set of files. User run applications, like a file syncing application, cannot be expected to efficiently manage files on the network.

To enable simple access to the network from the widest possible array of client applications, Storj implements a thin-client model that delegates trust to a dedicated server that manages data ownership. This is similar to the SPV wallet concept found in Bitcoin and other cryptocurrency ecosystems. The burdens of the data owner can be split across the client and the server in a variety of ways. By varying the amount of trust delegated, the server could also provide a wide variety of other valuable services. This sort of dedicated server, called Bridge, has been developed and released as Free Software. Any individual or organization can run their own Bridge server to facilitate network access.

Bridge is designed to store only metadata. It does not cache encrypted shards and, with the exception of public buckets, does not hold encryption keys. The only knowledge of the file that Bridge is able to share with third parties is metadata such as access patterns. This system protects the client's privacy and gives the client complete control over access to the data, while delegating the responsibility of keeping files available on the network to Bridge.

It is possible to envision Bridge upgrades that allow for different levels of delegated trust. A Bridge client may want to retain control over issuing and validating audits, or managing pointers to shards. Or a client may choose

to authorize two or more unrelated Bridges to manage its audits in order to minimize the trust it places in either Bridge server. In the long run, any function of the data owner can be split across two or more parties by delegating trust.

4.3 Client library

4.4 Bridge as a Network Information Repository

TODO As noted earlier, data owners are responsible for negotiating contracts and managing file state. With enough information about peers on the network, contract selection becomes a powerful tool for maintaining file state. A Bridge will have many active contracts with many farmers, and will therefore have access to information about those farmers. A Bridge could use this information to intelligently distribute shards across a set of farmers in order to achieve specific performance goals.

For instance, via the execution of a contract, a Bridge node gathers data about the farmer's communication latency, audit success rate, audit response latency, and availability. With minimal additional effort, the Bridge could also gather information about the node's available bandwidth. By gathering a large pool of reliable data about farmers, a Bridge node can intelligently select a set of farmers that collectively provides a probabilistic guarantee of a certain quality of service.

In other words, the Bridge can leverage its knowledge about peers on the network to tailor the service to the client's requirements. Rather than a limited set of service tiers, a Bridge could assemble a package of contracts on the fly to meet any service requirement. This allows the client to determine the optimal latency, bandwidth, or location of a file, and have confidence that its goals will be met. For instance, a streaming video application may specify a need for high bandwidth, while archival storage needs only high availability. In a sufficiently large network, any need could be met.

Secure distributed computation is an unsolved problem and, as such, each Bridge server uses its accumulated knowledge of the network. The Bridge is able to provide a probabilistic quality of service based on its knowledge the performance and reliability of farmers that a distributed network alone cannot provide.

4.5 Bridge as a Service

TODO In cases where the cost of delegating trust is not excessively high, clients may use third-party Bridges. Because Bridges do not store data and have no access to keys, this is still a large improvement on the traditional data-center model. Many of the features Bridge servers provide, like permissioning and intelligent contracting, leverage considerable network effects. Data sets grow exponentially more useful as they increase in size, indicating that there are strong economic incentives to share infrastructure and information in a Bridge.

Applications using object stores delegate significant amounts of trust to the storage providers. Providers may choose to operate public Bridges as a service. Application developers then delegate trust to the Bridge, as they would to a traditional object store, but to a lesser degree. Future updates will allow for various distributions of responsibilities (and thus levels of trust) between clients and Bridges. This shifts significant operational burdens from the application developer to the service-provider. This would also allow developers to pay for storage with standard payment mechanisms, like credit cards, rather than managing a cryptocurrency wallet. Storj Labs Inc. currently provides this service.

4.6 S3 gateway

TODO

5 Future Areas of Research

TODO Storj is a work in progress, and many features are planned for future versions. There are relatively few examples of functional distributed systems at scale, and many areas of research are still open.

5.1 Fast Byzantine Consensus

TODO

5.2 Distributed Repair

TODO

A Attacks

As with any distributed system, a variety of attack vectors exist. Many of these are common to all distributed systems. Some are storage-specific, and will apply to any distributed storage system.

A.1 Spartacus

TODO Spartacus attacks, or identity hijacking, are possible on Kademlia. Any node may assume the identity of another node and receive some fraction of messages intended for that node by simply copying its Node ID. This allows for targeted attacks against specific nodes and data. This is addressed by implementing Node IDs as ECDSA public key hashes and requiring messages be signed. A Spartacus attacker in this system would be unable to generate the corresponding private key, and thus unable to sign messages and participate in the network.

A.2 Sybil

Sybil attacks involve the creation of large amounts of nodes in an attempt to disrupt network operation by hijacking or dropping messages. Kademlia, because it relies on message redundancy and a concrete distance metric, is reasonably resistant to Sybil attacks. A node's neighbors in the network are selected by Node ID from an evenly distributed pool, and most messages are sent to at least three neighbors. If a Sybil attacker controls 50% of the network, it successfully isolates only 12.5% of honest nodes. While reliability and performance will degrade, the network will still be functional until a large portion of the network consists of colluding Sybil nodes.

A.2.1 Google

The Google attack, or nation-state attack, is a hypothetical variant of the Sybil attack carried out by an entity with extreme resources. Google attacks are hard to address, as it is difficult to predict the actions of an organization with orders of magnitude more resources than the sum of the resources of network participants. The only reliable defence against a Google attack is to create a network whose resources are on the same order of magnitude as the attacker's. At that scale, any attack against the network would represent an unsustainable

commitment of resources for such an organization.

A.2.2 Honest Geppetto

The Honest Geppetto attack is a storage-specific variant of the Google attack. The attacker operates a large number of ‘puppet’ nodes on the network, accumulating trust and contracts over time. Once he reaches a certain threshold he pulls the strings on each puppet to execute a hostage attack with the data involved, or simply drops each node from the network. Again, the best defence against this attack is to create a network of sufficient scale that this attack is ineffective. In the meantime, this can be partially addressed by relatedness analysis of nodes. Bayesian inference across downtime, latency and other attributes can be used to assess the likelihood that two nodes are operated by the same organization, and data owners can and should attempt to distribute shards across as many unrelated nodes as possible.

A.3 Eclipse

TODO mention S/Kademlia

An eclipse attack attempts to isolate a node or set of node in the network graph, by ensuring that all outbound connections reach malicious nodes. Eclipse attacks can be hard to identify, as malicious nodes can be made to function normally in most cases, only eclipsing certain important messages or information. Storj addresses eclipse attacks by using public key hashes as Node IDs. In order to eclipse any node in the network, the attacker must repeatedly generate key pairs until it finds three keys whose hashes are closer to the targeted node than its nearest non-malicious neighbor, and must defend that position against any new nodes with closer IDs. This is, in essence, a proof-of-work problem whose difficulty is proportional to the number of nodes in the network.

It follows that the best way to defend against eclipse attacks is to increase the number of nodes in the network. For large networks it becomes prohibitively expensive to perform an eclipse attack (see Section 6.2). Furthermore, any node that suspects it has been eclipsed may trivially generate a new keypair and node ID, thus restarting the proof-of-work challenge.

A.3.1 Tunnel Eclipse

TODO Because tunneled connections rely on the tunnel provider, it is trivial for a tunnel provider to eclipse nodes for which it provides tunneled connections. This attack cannot affect publicly addressable nodes, so it can be trivially defeated with proper configuration. This attack can be mitigated by encrypting messages intended for tunneled nodes, thus removing the malicious tunnel provider’s ability to inspect and censor incoming messages. Like a typical eclipse attack, any node that suspects it is the victim of a tunnel eclipse can easily generate a new Node ID, and find a new tunnel.

A.4 Hostage Bytes

The hostage byte attack is a storage-specific attack where malicious farmers refuse to transfer shards, or portions of shards, in order to extort additional payments from data owners. Data owners should protect themselves against hostage byte attacks by storing shards redundantly across several nodes (see Section 2.7). As long as the client keeps the bounds of its erasure encoding a secret, the malicious farmer cannot know what the last byte is. Redundant storage is not a complete solution for this attack, but addresses the vast majority of practical applications of this attack. Defeating redundancy requires collusion across multiple malicious nodes, which is difficult to execute in practice.

A.5 Cheating Owner

TODO A data owner may attempt to avoid paying a farmer for data storage by refusing to verify a correct audit. In response the farmer may drop the data-owner’s shard. This attack primarily poses a problem for any future distributed reputation system, as it is difficult for outside observers to verify the claims of either party. There is no known practical publicly verifiable proof of storage, and no known scheme for independently verifying that a privately verifiable audit was issued or answered as claimed. This indicates that a cheating client attack is a large unsolved problem for any reputation system.

A.6 Faithless Farmer

TODO While the farming software is built to require authentication via signature and token before serving download requests, it is reasonable to imagine a modification of the farming software that will provide shards to any paying requestor. In a network dominated by faithless farmers, any third-party can aggregate and inspect arbitrary shards present on the network.

However, even should faithless farmers dominate the network, data privacy is not significantly compromised. Because the location of the shards that comprise a given file is held solely by the data owner, it is prohibitively difficult to locate a target file without compromising the owner (see Section 6.3). Storj is not designed to protect against compromised data owners. In addition, should a third-party gather all shards, strong client-side encryption protects the contents of the file from inspection. The pointers and the encryption key may be secured separately. In the current implementation of Bridge, the pointers and the keys are held by the Bridge and the client, respectively.

A.7 Defeated Audit Attacks

TODO A typical Merkle proof verification does not require the verifier to know the depth of the tree. Instead the verifier is expected to have the data being validated. In the Storj audit tree, if the depth is unknown to the verifier the farmer may attack the verification process by sending a Merkle proof for any hash in the tree. This proof still generates the Merkle root, and is thus a valid proof of some node. But, because the verifier does not hold the data used to generate the tree, it has no way to verify that the proof is for the specific leaf that corresponds to the challenge. The verifier must store some information about the bottom of the tree, such as the depth of the tree, the set of leaves nodes, or the set of pre-leaves. Of these, the depth is most compact, and thus preferable.

Using the pre-leaf as an intermediary defeats another attack, where the farmer simply guesses which leaf corresponds to the current challenge. While this attack is unlikely to succeed, it's trivially defeated by forcing the farmer to provide the pre-leaf. The farmer cannot know the pre-leaf before the challenge is issued. Requiring transmission of the pre-leaf also allows the data owner to proceed through the challenge set linearly instead of being forced to select randomly. This is desirable because it allows the data owner to maintain less state information per tree.

B Selected Calculations

The following are several interesting calculations related to the operation of the network.

B.1 Difficulty of Eclipsing a Target Node

The probability of eclipsing a targeted node in the a network with k nodes in h hashes is modeled by a similar binomial distribution:

$$\Pr_{success}(h, k) = \sum_{i=3}^{h-1} k^{-i} \left(1 - \frac{1}{k}\right)^{h-i} \binom{h}{i}$$

h	i	$\Pr_{success} h, i$
100	100	7.937e-02
100	500	1.120e-03
100	900	2.046e-04
500	100	8.766e-01
500	500	8.012e-02
500	900	1.888e-02
900	100	9.939e-01
900	500	2.693e-01
900	900	8.020e-02

Code:

```

1 choose(h,k): return fac(h) / fac(k) / fac(h-k) def bin(i,h,k):
    return
2 choose(h,i) * k ** -i * (1-(1.0/k)) ** (h-i) def prob_succ(h,k):
    return
3 sum([bin(i,h,k) for i in range(3,h)])

```

B.2 Beach Size

As the number of shards on the network grows, it becomes progressively more difficult to locate a given file without prior knowledge of the locations of its shards. This implies that even should all farmers become faithless, file privacy is largely preserved.

The probability of locating a targeted file consisting of k shards by n random draws from a network containing N shards is modeled as a hypergeometric distribution with $K = k$:

$$Pr_{Success}(N, k, n) = \frac{\binom{N-k}{n-k} \binom{k}{n}}{\binom{N}{n}}$$

Code:

```

4 choose(h,k): return fac(h) / fac(k) / fac(h-k) def hyp(N,k,n):
    return

```

N	k	n	$Pr_{success} N, k, n$
100	10	10	5.777e-14
100	10	50	5.934e-04
100	10	90	3.305e-01
100	50	50	9.912e-30
100	50	90	5.493e-04
500	50	200	1.961e-22
500	50	400	7.361e-06
900	10	200	2.457e-07
900	10	400	2.823e-04
900	10	800	3.060e-01
900	50	200	1.072e-35
900	50	400	4.023e-19
900	50	800	2.320e-03

```

5 choose(N-k,n-k) / float(choose(N,n)) def prob_success(N,k,n):
    return hyp(N,k,n)

```

B.3 Partial Audit Confidence Levels

Farmers attempting to game the system may rely on data owners to issue partial audits. Partial audits allow false positives, where the data appears intact, but in fact has been modified. Data owners may account for this by ascribing confidence values to each partial audit, based on the likelihood of a false positive. Partial audit results then update prior confidence of availability. Data owners may adjust audit parameters to provide desired confidence levels.

The probability of a false positive on a partial audit of n bytes of an N byte shard, with K bytes modified adversarially by the farmer is a hypergeometric distribution with $k = 0$:

$$Pr_{falsepositive}(N, K, n) = \frac{\binom{N-K}{n}}{\binom{N}{n}}$$

Code:

```

6 choose(h,k): return fac(h) / fac(k) / fac(h-k) def hyp(N,K,n):
    return
7 float(choose(N-K, n) / choose(N,n)) def prob_false_pos(N,K,n):
    return hyp(N,K,n)

```

As demonstrated, the chance of false positives on even small partial audits

N	K	n	$\Pr_{falsepositive} N, K, n$
8192	512	512	1.466e-15
8192	1024	512	1.867e-31
8192	2048	512	3.989e-67
8192	3072	512	1.228e-109
8192	4096	512	2.952e-162

becomes vanishingly small. Farmers failing audits risk losing payouts from current contracts, as well as potential future contracts as a result of failed audits. Dropping 10% of a shard virtually guarantees a loss greater than 10% of the contract value. Thus it stands to reason that partially deleting shards to increase perceived storage capacity is not a viable economic strategy.

C Reed-Solomon

TODO

D Kademlia

TODO

E S/Kademlia

TODO

F Macaroons

TODO

References

- [1] Amitanand S. Aiyer, Lorenzo Alvisi, Allen Clement, Mike Dahlin, Jean-Philippe Martin, and Carl Porth. Bar fault tolerance for cooperative services. In *Proceedings of the Twentieth ACM Symposium on Operating Systems Principles*, SOSP '05, pages 45–58, New York, NY, USA, 2005. ACM.
- [2] Backblaze Inc. How long do hard drives last: 2018 hard drive stats. <https://www.backblaze.com/blog/hard-drive-stats-for-q1-2018/>, 2018.
- [3] Petar Maymounkov and David Mazières. Kademlia: A peer-to-peer information system based on the xor metric. In *Revised Papers from the First International Workshop on Peer-to-Peer Systems*, IPTPS '01, pages 53–65, London, UK, UK, 2002. Springer-Verlag.
- [4] Jeffrey Dean and Luiz André Barroso. The tail at scale. *Communications of the ACM*, 56:74–80, 2013.
- [5] Diego Ongaro and John Ousterhout. In search of an understandable consensus algorithm. In *Proceedings of the 2014 USENIX Conference on USENIX Annual Technical Conference*, USENIX ATC'14, pages 305–320, Berkeley, CA, USA, 2014. USENIX Association.
- [6] Google Inc. What is gRPC? <https://grpc.io/docs/guides/index.html>.
- [7] Arvid Norberg. uTorrent transport protocol. http://www.bittorrent.org/beps/bep_0029.html, 2009.
- [8] Ingmar Baumgart and Sebastian Mies. S/Kademlia: A practicable approach towards secure key-based routing. In *ICPADS*, pages 1–8. IEEE Computer Society, 2007.
- [9] Satoshi Nakamoto. Bitcoin: A peer-to-peer electronic cash system. <http://bitcoin.org/bitcoin.pdf>, 2008.
- [10] Amazon Inc. Amazon simple storage service - object metadata. <https://docs.aws.amazon.com/AmazonS3/latest/dev/UsingMetadata.html#object-metadata>.
- [11] B. Cohen. Incentives build robustness in bittorrent, (2003). <http://www.bittorrent.org/bittorrentecon.pdf>.
- [12] Jim Gray. Notes on data base operating systems. In *Operating Systems, An Advanced Course*, pages 393–481, London, UK, UK, 1978. Springer-Verlag.

- [13] E. A. Akkoyunlu, K. Ekanadham, and R. V. Huber. Some constraints and tradeoffs in the design of network communications. In *Proceedings of the Fifth ACM Symposium on Operating Systems Principles*, SOSP '75, pages 67–74, New York, NY, USA, 1975. ACM.
- [14] Seth Gilbert and Nancy Lynch. Perspectives on the cap theorem. *Computer*, 45(2):30–36, February 2012.
- [15] Kyle Kingsbury. Strong consistency models. <https://aphyr.com/posts/313-strong-consistency-models>, 2014.
- [16] Justin Sheehy. There is no now. *Queue*, 13(3):20:20–20:27, March 2015.
- [17] Amazon Inc. Amazon simple storage service - data consistency model. <https://docs.aws.amazon.com/AmazonS3/latest/dev/Introduction.html#ConsistencyModel>.
- [18] Giuseppe DeCandia, Deniz Hastorun, Madan Jampani, Gunavardhan Kakulapati, Avinash Lakshman, Alex Pilchin, Swaminathan Sivasubramanian, Peter Voshall, and Werner Vogels. Dynamo: Amazon’s highly available key-value store. In *Proceedings of Twenty-first ACM SIGOPS Symposium on Operating Systems Principles*, SOSP '07, pages 205–220, New York, NY, USA, 2007. ACM.
- [19] Avinash Lakshman and Prashant Malik. Cassandra: A decentralized structured storage system. *SIGOPS Oper. Syst. Rev.*, 44(2):35–40, April 2010.
- [20] Michael J. Fischer, Nancy A. Lynch, and Michael S. Paterson. Impossibility of distributed consensus with one faulty process. *J. ACM*, 32(2):374–382, April 1985.
- [21] Brian M. Oki and Barbara H. Liskov. Viewstamped replication: A new primary copy method to support highly-available distributed systems. In *Proceedings of the Seventh Annual ACM Symposium on Principles of Distributed Computing*, PODC '88, pages 8–17, New York, NY, USA, 1988. ACM.
- [22] Leslie Lamport. The part-time parliament website note. <https://www.microsoft.com/en-us/research/publication/part-time-parliament/>.
- [23] Leslie Lamport. The part-time parliament. *ACM Trans. Comput. Syst.*, 16(2):133–169, May 1998.
- [24] Leslie Lamport. Paxos made simple. pages 51–58, December 2001.

- [25] Tushar Deepak Chandra, Robert Griesemer, and Joshua Redstone. Paxos made live - an engineering perspective (2006 invited talk). In *Proceedings of the 26th Annual ACM Symposium on Principles of Distributed Computing*, 2007.
- [26] Robbert Van Renesse and Deniz Altinbuken. Paxos made moderately complex. *ACM Comput. Surv.*, 47(3):42:1–42:36, February 2015.
- [27] Barbara Liskov and James Cowling. Viewstamped replication revisited. Technical Report MIT-CSAIL-TR-2012-021, MIT, July 2012.
- [28] Robbert van Renesse and Fred B. Schneider. Chain replication for supporting high throughput and availability. In *Proceedings of the 6th Conference on Symposium on Operating Systems Design & Implementation - Volume 6*, OSDI'04, pages 7–7, Berkeley, CA, USA, 2004. USENIX Association.
- [29] Flavio Paiva Junqueira, Benjamin C. Reed, and Marco Serafini. Zab: High-performance broadcast for primary-backup systems. *2011 IEEE/IFIP 41st International Conference on Dependable Systems & Networks (DSN)*, pages 245–256, 2011.
- [30] Iulian Moraru, David G. Andersen, and Michael Kaminsky. There is more consensus in egalitarian parliaments. In *Proceedings of the Twenty-Fourth ACM Symposium on Operating Systems Principles*, SOSP '13, pages 358–372, New York, NY, USA, 2013. ACM.
- [31] H. Howard, D. Malkhi, and A. Spiegelman. Flexible Paxos: Quorum intersection revisited. *ArXiv e-prints*, August 2016.
- [32] Mike Burrows. The chubby lock service for loosely-coupled distributed systems. In *Proceedings of the 7th Symposium on Operating Systems Design and Implementation*, OSDI '06, pages 335–350, Berkeley, CA, USA, 2006. USENIX Association.
- [33] Fay Chang, Jeffrey Dean, Sanjay Ghemawat, Wilson C. Hsieh, Deborah A. Wallach, Mike Burrows, Tushar Chandra, Andrew Fikes, and Robert E. Gruber. Bigtable: A distributed storage system for structured data. In *7th USENIX Symposium on Operating Systems Design and Implementation (OSDI)*, pages 205–218, 2006.
- [34] James C. Corbett, Jeffrey Dean, Michael Epstein, Andrew Fikes, Christopher Frost, JJ Furman, Sanjay Ghemawat, Andrey Gubarev, Christopher Heiser, Peter Hochschild, Wilson Hsieh, Sebastian Kanthak, Eugene Kogan, Hongyi Li, Alexander Lloyd, Sergey Melnik, David Mwaura, David Nagle, Sean Quinlan, Rajesh Rao, Lindsay Rolig, Dale

- Woodford, Yasushi Saito, Christopher Taylor, Michal Szymaniak, and Ruth Wang. Spanner: Google’s globally-distributed database. In *OSDI*, 2012.
- [35] Miguel Castro and Barbara Liskov. Practical byzantine fault tolerance. In *Proceedings of the Third Symposium on Operating Systems Design and Implementation*, OSDI ’99, pages 173–186, Berkeley, CA, USA, 1999. USENIX Association.
- [36] Michael Abd-El-Malek, Gregory R. Ganger, Garth R. Goodson, Michael K. Reiter, and Jay J. Wylie. Fault-scalable byzantine fault-tolerant services. In *Proceedings of the Twentieth ACM Symposium on Operating Systems Principles*, SOSP ’05, pages 59–74, New York, NY, USA, 2005. ACM.
- [37] Jean-Philippe Martin and Lorenzo Alvisi. Fast byzantine consensus. *IEEE Trans. Dependable Secur. Comput.*, 3(3):202–215, July 2006.
- [38] I. Abraham, G. Gueta, D. Malkhi, L. Alvisi, R. Kotla, and J.-P. Martin. Revisiting Fast Practical Byzantine Fault Tolerance. *ArXiv e-prints*, December 2017.
- [39] Ramakrishna Kotla. Zyzzyva: Speculative byzantine fault tolerance. *ACM Transactions on Computer Systems (TOCS)*, 27, Issue 4, Article No. 7, December 2009.
- [40] P. L. Aublin, S. B. Mokhtar, and V. Quéma. RBFT: Redundant Byzantine Fault Tolerance. In *2013 IEEE 33rd International Conference on Distributed Computing Systems*, pages 297–306, July 2013.
- [41] Christopher N. Copeland and Hongxia Zhong. Tangaroa: a Byzantine Fault Tolerant Raft, 2014.
- [42] Jae Kwon. Tendermint: Consensus without mining. <https://tendermint.com/docs/tendermint.pdf>, 2014.
- [43] Pierre-Louis Aublin, Rachid Guerraoui, Nikola Knežević, Vivien Quéma, and Marko Vukolić. The next 700 bft protocols. *ACM Trans. Comput. Syst.*, 32(4):12:1–12:45, January 2015.
- [44] Leemon Baird. The swirlds hashgraph consensus algorithm: Fair, fast, byzantine fault tolerance. 2016.
- [45] Andrew Miller, Yu Xia, Kyle Croman, Elaine Shi, and Dawn Song. The Honey Badger of BFT Protocols. Cryptology ePrint Archive, Report 2016/199, 2016. <https://eprint.iacr.org/2016/199>.

- [46] Yossi Gilad, Rotem Hemo, Silvio Micali, Georgios Vlachos, and Nickolai Zeldovich. Algorand: Scaling byzantine agreements for cryptocurrencies. In *Proceedings of the 26th Symposium on Operating Systems Principles*, SOSP '17, pages 51–68, New York, NY, USA, 2017. ACM.
- [47] Vitalik Buterin and Virgil Griffith. Casper the friendly finality gadget. *CoRR*, abs/1710.09437, 2017.
- [48] Serguei Popov. The tangle. https://iota.org/IOTA_Whitepaper.pdf, 2018.
- [49] Team Rocket. Snowflake to Avalanche: A Novel Metastable Consensus Protocol Family for Cryptocurrencies. <https://ipfs.io/ipfs/QmUy4jh5mGNZvLkjies1RWM4YuvJh5o2FYopNPVYwrRVGV>, 2018.
- [50] Pierre Chevalier, Bartłomiej Kamiński, Fraser Hutchison, Qi Ma, and Spandan Sharma. Protocol for Asynchronous, Reliable, Secure and Efficient Consensus (PARSEC). <http://docs.maidsafe.net/Whitepapers/pdf/PARSEC.pdf>, 2018.
- [51] James Mickens. The saddest moment. ;login: logout, May 2013. <https://scholar.harvard.edu/files/mickens/files/thesaddestmoment.pdf>.
- [52] Matteo Zignani, Sabrina Gaito, and Gian Paolo Rossi. Follow the "mastodon": Structure and evolution of a decentralized online social network, 2018.