Chapter 10

Peer-to-Peer Computing

"Indeed, I believe that virtually *every* important aspect of programming arises somewhere in the context of [sorting and] searching!"

- Donald E. Knuth, The Art of Computer Programming

10.1 Introduction

Unfortunately, the term *peer-to-peer* (P2P) is ambiguous, used in a variety of different contexts, such as:

- In popular media coverage, P2P is often synonymous to software or protocols that allow users to "share" files, often of dubious origin. In the early days, P2P users mostly shared music, pictures, and software; nowadays books, movies or tv shows have caught on. P2P file sharing is immensely popular, currently at least half of the total Internet traffic is due to P2P!
- In academia, the term P2P is used mostly in two ways. A narrow view essentially defines P2P as the "theory behind file sharing protocols". In other words, how do Internet hosts need to be organized in order to deliver a search engine to find (file sharing) content efficiently? A popular term is "distributed hash table" (DHT), a distributed data structure that implements such a content search engine. A DHT should support at least a search (for a key) and an insert (key, object) operation. A DHT has many applications beyond file sharing, e.g., the Internet domain name system (DNS).
- A broader view generalizes P2P beyond file sharing: Indeed, there is a growing number of applications operating outside the juridical gray area, e.g., P2P Internet telephony à la Skype, P2P mass player games on video consoles connected to the Internet, or P2P live video streaming as in Zattoo. So, again, what is P2P?! Still not an easy question... Trying to account for the new applications beyond file sharing, one might define P2P as a large-scale distributed system that operates without a central server bottleneck. However, with this definition almost everything we

learn in this course is P2P! Moreover, according to this definition earlyday file sharing applications such as Napster (1999) that essentially made the term P2P popular would not be P2P! On the other hand, the plain old telephone system or the world wide web do fit the P2P definition...

• From a different viewpoint, the term P2P may also be synonymous for privacy protection, as various P2P systems such as Freenet allow publishers of information to remain anonymous and uncensored. (Studies show that these freedom-of-speech P2P networks do not feature a lot of content against oppressive governments; indeed the majority of text documents seem to be about illicit drugs, not to speak about the type of content in audio or video files.)

In other words, we cannot hope for a single well-fitting definition of P2P, as some of them even contradict. In the following we mostly employ the academic viewpoints (second and third definition above). In this context, it is generally believed that P2P will have an influence on the future of the Internet. The P2P paradigm promises to give better scalability, availability, reliability, privacy, and security, just about everything researchers expect from a future Internet architecture.

One might naively assume that for instance scalability is not an issue in today's Internet, as even most popular web pages are generally highly available. However, this is not really because of our well-designed Internet architecture, but rather due to the help of so-called overlay networks: The Google website for instance manages to respond so reliably and quickly because Google maintains a large distributed infrastructure, essentially a P2P system. Similarly companies like Akamai sell "P2P functionality" to their customers to make today's user experience possible in the first place. Quite possibly today's P2P applications are just testbeds for tomorrow's Internet architecture.

10.2 Architecture Variants

Several P2P architectures are known:

- Client/Server goes P2P: Even though Napster is known to the be first P2P system (1999), by today's standards its architecture would not deserve the label P2P anymore. Napster clients accessed a central server that managed all the information of the shared files, i.e., which file was to be found on which client. Only the downloading process itself was between clients ("peers") directly, hence peer-to-peer. In the early days of Napster the load of the server was relatively small, so the simple Napster architecture made a lot of sense. Later on, it became clear that the server would eventually be a bottleneck, and more so an attractive target for an attack. Indeed, eventually a judge ruled the server to be shut down, in other words, he conducted a juridical denial of service attack.
- Unstructured P2P: The Gnutella protocol is the anti-thesis of Napster, as it is a fully decentralized system, with no single entity having a global picture. Instead each peer would connect to a random sample of other peers, constantly changing the neighbors of this virtual overlay network by exchanging neighbors with neighbors of neighbors. (In such a system

it is part of the challenge to find a decentralized way to even discover a first neighbor; this is known as the bootstrap problem. To solve it, usually some random peers of a list of well-known peers are contacted first.) When searching for a file, the request was being flooded in the network (Algorithm 11 in Chapter 3). Indeed, since users often turn off their client once they downloaded their content there usually is a lot of *churn* (peers joining and leaving at high rates) in a P2P system, so selecting the right "random" neighbors is an interesting research problem by itself. However, unstructured P2P architectures such as Gnutella have a major disadvantage, namely that each search will cost m messages, m being the number of virtual edges in the architecture. In other words, such an unstructured P2P architecture will not scale.

- Hybrid P2P: The synthesis of client/server architectures such as Napster and unstructured architectures such as Gnutella are hybrid architectures. Some powerful peers are promoted to so-called superpeers (or, similarly, trackers). The set of superpeers may change over time, and taking down a fraction of superpeers will not harm the system. Search requests are handled on the superpeer level, resulting in much less messages than in flat/homogeneous unstructured systems. Essentially the superpeers together provide a more fault-tolerant version of the Napster server, all regular peers connect to a superpeer. As of today, almost all popular P2P systems have such a hybrid architecture, carefully trading off reliability and efficiency, but essentially not using any fancy algorithms and techniques.
- Structured P2P: Inspired by the early success of Napster, the academic world started to look into the question of efficient file sharing. Indeed, even earlier, in 1997, Plaxton, Rajaraman, and Richa proposed a hypercubic architecture for P2P systems. This was a blueprint for many so-called structured P2P architecture proposals, such as Chord, CAN, Pastry, Tapestry, Viceroy, Kademlia, Koorde, SkipGraph, SkipNet, etc. In practice structured P2P architectures are not yet popular, apart from the Kad (from Kademlia) architecture which comes for free with the eMule client. Indeed, also the Plaxton et al. paper was standing on the shoulders of giants. Some of its eminent precursors are:
 - Research on linear and consistent hashing, e.g., the paper "Consistent hashing and random trees: Distributed caching protocols for relieving hot spots on the World Wide Web" by Karger et al. (co-authored also by the late Daniel Lewin from Akamai), 1997.
 - Research on locating shared objects, e.g., the papers "Sparse Partitions" (see also Chapter 6) or "Concurrent Online Tracking of Mobile Users" by Awerbuch and Peleg, 1990 and 1991.
 - Work on so-called compact routing: The idea is to construct routing tables such that there is a trade-off between memory (size of routing tables) and stretch (quality of routes), e.g., "A trade-off between space and efficiency for routing tables" by Peleg and Upfal, 1988.
 - $-\ldots$ and even earlier: hypercubic networks, see next section!

10.3 Hypercubic Networks

(Thanks to Christian Scheideler, TUM, for the pictures in this section.)

In this section we will introduce some popular families of network topologies. These topologies are used in countless application domains, e.g., in classic parallel computers or telecommunication networks, or more recently (as said above) in the emerging area of P2P computing. Similarly to Chapter 9 we employ the All-to-All communication model of Chapter 7, i.e., each node can set up direct communication links to arbitrary other nodes. Such a virtual network is called an *overlay topology*, or in this context, P2P architecture. In this section we present a few overlay topologies of general interest.

The most basic network topologies used in practice are trees, rings, grids or tori. Many other suggested networks are simply combinations or derivatives of these. The advantage of trees is that the routing is very easy: for every source-destination pair there is only one possible simple path. However, since the root of a tree is usually a severe bottleneck, so-called *fat trees* have been used. These trees have the property that every edge connecting a node v to its parent u has a capacity that is equal to all leaves of the subtree routed at v. See Figure 10.1 for an example.



Figure 10.1: The structure of a fat tree.

Remark:

• Fat trees belong to a family of networks that require edges of non-uniform capacity to be efficient. Easier to build are networks with edges of uniform capacity. This is usually the case for grids and tori. Unless explicitly mentioned, we will treat all edges in the following to be of capacity 1. In the following, [x] means the set $\{0, 1, \ldots, x-1\}$.

Definition 10.1 (Torus, Mesh). Let $m, d \in N$. The (m, d)-mesh M(m, d) is a graph with node set $V = [m]^d$ and edge set

$$E = \left\{ \{ (a_{d-1} \dots a_0), (b_{d-1} \dots b_0) \} \mid a_i, b_i \in [m], \sum_{i=0}^{d-1} |a_i - b_i| = 1 \right\} .$$

The (m,d)-torus T(m,d) is a graph that consists of an (m,d)-mesh and additionally wrap-around edges from $(a_{d-1} \dots a_{i+1}(m-1) \ a_{i-1} \dots a_0)$ to $(a_{d-1} \dots a_{i+1} \ 0 \ a_{i-1} \dots a_0)$ for all $i \in [d]$ and all $a_j \in [m]$ with $j \neq i$. M(m,1)is also called a line, T(m,1) a cycle, and M(2,d) = T(2,d) a d-dimensional hypercube. Figure 10.2 presents a linear array, a torus, and a hypercube.



Figure 10.2: The structure of M(m, 1), T(4, 2), and M(2, 3).

Remarks:

- Routing on mesh, torus, and hypercube is trivial. On a *d*-dimensional hypercube, to get from a source bitstring *s* to a target bitstring *d* one only needs to fix each "wrong" bit, one at a time; in other words, if the source and the target differ by *k* bits, there are *k*! routes with *k* hops.
- The hypercube can directly be used for a structured P2P architecture. It is trivial to construct a distributed hash table (DHT): We have *n* nodes, *n* for simplicity being a power of 2, i.e., $n = 2^d$. As in the hypercube, each node gets a unique *d*-bit ID, and each node connects to *d* other nodes, i.e., the nodes that have IDs differing in exactly one bit. Now we use a globally known hash function f(), mapping file names to long bit strings; SHA-1 is popular in practice, providing 160 bits. Let $f_d()$ be the first *d* bits (prefix) of the bitstring produced by f(). If a node is searching for file name X, it routes a request message f(X) to node $f_d(X)$. Clearly, node $f_d(X)$ can only answer this request if all files with hash prefix $f_d(X)$ have been previously registered at node $f_d(X)$.
- There are a few issues which need to be addressed before our DHT works, in particular churn (nodes joining and leaving without notice). To deal with churn the system needs some level of replication, i.e., a number of nodes which are responsible for each prefix such that failure of some nodes will not compromise the system. We give some more details in Section 10.4. In addition there are other issues (e.g., security, efficiency) which can be addressed to improve the system. Delay efficiency for instance is already considered in the seminal paper by Plaxton et al. These issues are beyond the scope of this lecture.
- The hypercube has many derivatives, the so-called *hypercubic networks*. Among these are the butterfly, cube-connected-cycles, shuffle-exchange, and de Bruijn graph. We start with the butterfly, which is basically a "rolled out" hypercube (hence directly providing replication!).

Definition 10.2 (Butterfly). Let $d \in N$. The d-dimensional butterfly BF(d) is a graph with node set $V = [d+1] \times [2]^d$ and an edge set $E = E_1 \cup E_2$ with

$$E_1 = \{\{(i, \alpha), (i+1, \alpha)\} \mid i \in [d], \ \alpha \in [2]^d\}$$

and

$$E_2 = \{\{(i,\alpha), (i+1,\beta)\} \mid i \in [d], \ \alpha, \beta \in [2]^d, \ \alpha \text{ and } \beta \text{ differ} \\ only \text{ at the } i^{th} \text{ position}\} .$$

A node set $\{(i, \alpha) \mid \alpha \in [2]^d\}$ is said to form level *i* of the butterfly. The *d*-dimensional wrap-around butterfly *W*-*BF*(*d*) is defined by taking the *BF*(*d*) and identifying level *d* with level 0.

Remarks:

- Figure 10.3 shows the 3-dimensional butterfly BF(3). The BF(d) has $(d+1)2^d$ nodes, $2d \cdot 2^d$ edges and degree 4. It is not difficult to check that combining the node sets $\{(i, \alpha) \mid i \in [d]\}$ into a single node results in the hypercube.
- Butterflies have the advantage of a constant node degree over hypercubes, whereas hypercubes feature more fault-tolerant routing.
- The structure of a butterfly might remind you of sorting networks from Chapter 9. Although butterflies are used in the P2P context (e.g. Viceroy), they have been used decades earlier for communication switches. The well-known Benes network is nothing but two back-to-back butterflies. And indeed, butterflies (and other hypercubic networks) are even older than that; students familiar with fast fourier transform (FFT) will recognize the structure without doubt. Every year there is a new application for which a hypercubic network is the perfect solution!
- Indeed, hypercubic networks are related. Since all structured P2P architectures are based on hypercubic networks, they in turn are all related.
- Next we define the cube-connected-cycles network. It only has a degree of 3 and it results from the hypercube by replacing the corners by cycles.



Figure 10.3: The structure of BF(3).

Definition 10.3 (Cube-Connected-Cycles). Let $d \in N$. The cube-connected-cycles network CCC(d) is a graph with node set $V = \{(a, p) \mid a \in [2]^d, p \in [d]\}$ and edge set

$$E = \{\{(a,p), (a, (p+1) \mod d)\} \mid a \in [2]^d, p \in [d]\} \\ \cup \{\{(a,p), (b,p)\} \mid a, b \in [2]^d, p \in [d], a = b \text{ except for } a_p\}$$



Figure 10.4: TThe structure of CCC(3).

Remarks:

- Two possible representations of a CCC can be found in Figure 10.4.
- The shuffle-exchange is yet another way of transforming the hypercubic interconnection structure into a constant degree network.

Definition 10.4 (Shuffle-Exchange). Let $d \in N$. The d-dimensional shuffleexchange SE(d) is defined as an undirected graph with node set $V = [2]^d$ and an edge set $E = E_1 \cup E_2$ with

$$E_1 = \{\{(a_{d-1}\dots a_0), (a_{d-1}\dots \bar{a}_0)\} \mid (a_{d-1}\dots a_0) \in [2]^d, \ \bar{a}_0 = 1 - a_0\}$$

and

$$E_2 = \{\{(a_{d-1} \dots a_0), (a_0 a_{d-1} \dots a_1)\} \mid (a_{d-1} \dots a_0) \in [2]^d\} .$$

Figure 10.5 shows the 3- and 4-dimensional shuffle-exchange graph.



Figure 10.5: The structure of SE(3) and SE(4).

Definition 10.5 (DeBruijn). The b-ary DeBruijn graph of dimension d DB(b,d) is an undirected graph G = (V, E) with node set $V = \{v \in [b]^d\}$ and edge set E that contains all edges $\{v, w\}$ with the property that $w \in \{(x, v_{d-1}, \ldots, v_1) : x \in [b]\}$, where $v = (v_{d-1}, \ldots, v_0)$.



Figure 10.6: The structure of DB(2,2) and DB(2,3).

Remarks:

- Two examples of a DeBruijn graph can be found in Figure 10.6. The DeBruijn graph is the basis of the Koorde P2P architecture.
- There are some data structures which also qualify as hypercubic networks. An obvious example is the Chord P2P architecture, which uses a slightly different hypercubic topology. A less obvious (and therefore good) example is the skip list, the balanced binary search tree for the lazy programmer:

Definition 10.6 (Skip List). The skip list is an ordinary ordered linked list of objects, augmented with additional forward links. The ordinary linked list is the level 0 of the skip list. In addition, every object is promoted to level 1 with probability 1/2. As for level 0, all level 1 objects are connected by a linked list. In general, every object on level i is promoted to the next level with probability 1/2. A special start-object points to the smallest/first object on each level.

Remark:

- Search, insert, and delete can be implemented in $O(\log n)$ expected time in a skip list, simply by jumping from higher levels to lower ones when overshooting the searched position. Also, the amortized memory cost of each object is constant, as on average an object only has two forward pointers.
- The randomization can easily be discarded, by deterministically promoting a constant fraction of objects of level i to level i + 1, for all i. When inserting or deleting, object o simply checks whether its left and right level i neighbors are being promoted to level i + 1. If none of them is, promote object o itself. Essentially we establish a MIS on each level, hence at least every third and at most every second object is promoted.
- There are obvious variants of the skip list, e.g., the skip graph. Instead of promoting only half of the nodes to the next level, we always promote all the nodes, similarly to a balanced binary tree: All nodes are part of the root level of the binary tree. Half the nodes are promoted left, and half the nodes are promoted right, on each level. Hence on level *i* we have

have 2^i lists (or, more symmetrically: rings) of about $n/2^i$ objects. This is pretty much what we need for a nice hypercubic P2P architecture.

• One important goal in choosing a topology for a network is that it has a small diameter. The following theorem presents a lower bound for this.

Theorem 10.7. Every graph of maximum degree d > 2 and size n must have a diameter of at least $|(\log n)/(\log(d-1))| - 1$.

Proof. Suppose we have a graph G = (V, E) of maximum degree d and size n. Start from any node $v \in V$. In a first step at most d other nodes can be reached. In two steps at most $d \cdot (d-1)$ additional nodes can be reached. Thus, in general, in at most k steps at most

$$1 + \sum_{i=0}^{k-1} d \cdot (d-1)^i = 1 + d \cdot \frac{(d-1)^k - 1}{(d-1) - 1} \le \frac{d \cdot (d-1)^k}{d-2}$$

nodes (including v) can be reached. This has to be at least n to ensure that v can reach all other nodes in V within k steps. Hence,

$$(d-1)^k \ge \frac{(d-2)\cdot n}{d} \quad \Leftrightarrow \quad k \ge \log_{d-1}((d-2)\cdot n/d) .$$

Since $\log_{d-1}((d-2)/d) > -2$ for all d > 2, this is true only if $k \ge \lfloor \log_{d-1} n \rfloor - 1$.

Remarks:

- In other words, constant-degree hypercubic networks feature an asymptotically optimal diameter.
- There are a few other interesting graph classes, e.g., expander graphs (an expander graph is a sparse graph which has high connectivity properties, that is, from every not too large subset of nodes you are connected to a larger set of nodes), or small-world graphs (popular representations of social networks). At first sight hypercubic networks seem to be related to expanders and small-world graphs, but they are not.

10.4 DHT & Churn

As written earlier, a DHT essentially is a hypercubic structure with nodes having identifiers such that they span the ID space of the objects to be stored. We described the straightforward way how the ID space is mapped onto the peers for the hypercube. Other hypercubic structures may be more complicated: The butterfly network, for instance, may directly use the d + 1 layers for replication, i.e., all the d + 1 nodes with the same ID are responsible for the same hash prefix. For other hypercubic networks, e.g., the pancake graph (see exercises), assigning the object space to peer nodes may be more difficult.

In general a DHT has to withstand churn. Usually, peers are under control of individual users who turn their machines on or off at any time. Such peers join and leave the P2P system at high rates ("churn"), a problem that is not existent in orthodox distributed systems, hence P2P systems fundamentally differ from old-school distributed systems where it is assumed that the nodes in the system are relatively stable. In traditional distributed systems a single unavailable node is a minor disaster: all the other nodes have to get a consistent view of the system again, essentially they have to reach consensus which nodes are available. In a P2P system there is usually so much churn that it is impossible to have a consistent view at any time.

Most P2P systems in the literature are analyzed against an adversary that can crash a fraction of random peers. After crashing a few peers the system is given sufficient time to recover again. However, this seems unrealistic. The scheme sketched in this section significantly differs from this in two major aspects. First, we assume that joins and leaves occur in a worst-case manner. We think of an adversary that can remove and add a bounded number of peers; it can choose which peers to crash and how peers join. We assume that a joining peer knows a peer which already belongs to the system. Second, the adversary does not have to wait until the system is recovered before it crashes the next batch of peers. Instead, the adversary can constantly crash peers, while the system is trying to stay alive. Indeed, the system is never fully repaired but always fully functional. In particular, the system is resilient against an adversary that continuously attacks the "weakest part" of the system. The adversary could for example insert a crawler into the P2P system, learn the topology of the system, and then repeatedly crash selected peers, in an attempt to partition the P2P network. The system counters such an adversary by continuously moving the remaining or newly joining peers towards the sparse areas.

Clearly, we cannot allow the adversary to have unbounded capabilities. In particular, in any constant time interval, the adversary can at most add and/or remove $O(\log n)$ peers, n being the total number of peers currently in the system. This model covers an adversary which repeatedly takes down machines by a distributed denial of service attack, however only a logarithmic number of machines at each point in time. The algorithm relies on messages being delivered timely, in at most constant time between any pair of operational peers, i.e., the synchronous model. Using the trivial synchronizer this is not a problem. We only need bounded message delays in order to have a notion of time which is needed for the adversarial model. The duration of a round is then proportional to the propagation delay of the slowest message.

In the remainder of this section, we give a sketch of the system: For simplicity, the basic structure of the P2P system is a hypercube. Each peer is part of a distinct hypercube node; each hypercube node consists of $\Theta(\log n)$ peers. Peers have connections to other peers of their hypercube node and to peers of the neighboring hypercube nodes.¹ Because of churn, some of the peers have to change to another hypercube node such that up to constant factors, all hypercube nodes own the same number of peers at all times. If the total number of peers grows or shrinks above or below a certain threshold, the dimension of the hypercube is increased or decreased by one, respectively.

The balancing of peers among the hypercube nodes can be seen as a dynamic token distribution problem on the hypercube. Each node of the hypercube has a certain number of tokens, the goal is to distribute the tokens along the edges of the graph such that all nodes end up with the same or almost the same number

¹Having a logarithmic number of hypercube neighbor nodes, each with a logarithmic number of peers, means that each peers has $\Theta(\log^2 n)$ neighbor peers. However, with some additional bells and whistles one can achieve $\Theta(\log n)$ neighbor peers.

of tokens. While tokens are moved around, an adversary constantly inserts and deletes tokens. See also Figure 10.7.



Figure 10.7: A simulated 2-dimensional hypercube with four nodes, each consisting of several peers. Also, all the peers are either in the core or in the periphery of a node. All peers within the same node are completely connected to each other, and additionally, all peers of a node are connected to the core peers of the neighboring nodes. Only the core peers store data items, while the peripheral peers move between the nodes to balance biased adversarial changes.

In summary, the P2P system builds on two basic components: i) an algorithm which performs the described dynamic token distribution and ii) an information aggregation algorithm which is used to estimate the number of peers in the system and to adapt the dimension of the hypercube accordingly:

Theorem 10.8 (DHT with Churn). We have a fully scalable, efficient P2P system which tolerates $O(\log n)$ worst-case joins and/or crashes per constant time interval. As in other P2P systems, peers have $O(\log n)$ neighbors, and the usual operations (e.g., search, insert) take time $O(\log n)$.

Remarks:

- Indeed, handling churn is only a minimal requirement to make a P2P system work. Later studies proposed more elaborate architectures which can also handle other security issues, e.g., privacy or Byzantine attacks.
- It is surprising that unstructured (in fact, hybrid) P2P systems dominate structured P2P systems in the real world. One would think that structured P2P systems have advantages, in particular their efficient logarithmic data lookup. On the other hand, unstructured P2P networks are simpler, in particular in light of non-exact queries.