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On-Demand Media Streaming Over the Internet

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Abstract

We propose a new model for on-demand media streaming centered around the peer-to-peer (P2P) paradigm. The proposed P2P model can support a large number of clients with a low overall system cost. The P2P model allows for peers to share some of their resources with the system and in return, they get some incentives or rewards. We describe how to realize (or deploy) the proposed model. In addition, we present a new dispersion algorithm (for disseminating the media files into the system) and a searching algorithm (for locating peers with the required objects).

We demonstrate the potential of the P2P model as an infrastructure for a large-scale on-demand media streaming service through an extensive simulation study on large, Internet-like, topologies. Starting with a limited streaming capacity (hence, low cost), the simulation shows that the capacity is rapidly increased and many clients can be served even if they come according to different arrival patterns such as constant rate arrivals, flash crowd arrivals, and Poisson arrivals.

1. Introduction

Streaming multimedia files to a large number of customers imposes a high load on the underlying network and the streaming server. The voluminous nature of the multimedia traffic along with its timing constraints make deploying a large-scale, cost effective, media streaming architecture over the current Internet a challenge.

The current media streaming architectures are mainly composed of a streaming entity and a set of requesting clients. The supplying entity could be one server, a set of servers, a set of servers and caches, or a set of servers and proxies. This entity is responsible for providing the requested media files to *all* clients. The total number of concurrent clients the system can support, called the overall system capacity, is limited by the resources of the stream-

ing entity. The limitation mainly comes from the out bound network bandwidth, but it could also be due to the processing power, memory size, or the I/O speed of the server machine. For instance, a streaming server hooked to the Internet through a T3 link (\sim 45 Mb/s) would be able to support up to 45 concurrent users requesting a media file recorded at 1 Mb/s. These approaches have limitations in reliability and scalability. The reliability concern arises from the fact that only one entity is feeding all clients, i.e., a single point of failure. The scalability of these approaches is not on a par with the requirements of a media distribution service that spans Internet-scale potential users, since adding more users requires adding a commensurate amount of resources to the supplying server.

Whereas deploying proxies and caches at several locations over the Internet increases the overall system capacity, it multiplies the overall system cost and introduces many administrative challenges such as cache consistency and load balancing problems. The system capacity is still limited by the aggregate resources of the caches and proxies. This shifts the bottleneck from one central point to a "few" distributed points, but it does not eliminate it.

We propose a peer-to-peer (P2P) media distribution model that can support a large number of clients with a low overall system cost. The key idea of the model is that peers *share* some of their resources with the system. In return, they get *incentives* or rewards from the service provider. As peers contribute resources to the system, the capacity increases and more clients can be served. By properly motivating peers, the service provider can achieve a large system capacity with a relatively small initial investment. A peer-to-peer architecture has the potential to provide the desired large-scale media distribution service.

There is a difference between a file-sharing system and a media streaming system [12]. In file-sharing systems, a client first downloads the *entire* file before using it. The shared files are typically small (few Mbytes) and take a relatively short time to download. A file is stored entirely by one peer and hence, a requesting peer needs to establish



only one connection to download it. There are no timing constraints on downloading the fragments of the file, rather the total download time is more important. This means that the system can tolerate inter-packet delays. In media streaming systems, a client *overlaps* downloading with the consumption of the file. It uses one part while downloading another to be used in the immediate future. The files are large (on the order of Gbytes) and take long time to stream. A large media file is expected to be stored by several peers, which requires the requesting peer to manage several connections concurrently. Finally, timing constraints are crucial, since a packet arriving after its scheduled play back time is useless and considered lost.

The remainder of the paper is organized as follows. Section 2 presents the P2P model. Section 3 presents the protocol to be run by a participating peer in the system. The architecture along with the searching and dispersion algorithms are presented in Section 4. The simulation study is presented in Section 5. Section 6 summarizes the related research effort. Section 7 concludes the paper and proposes future extensions for this research.

2. P2P Model for Media Streaming

In the P2P model, a peer may act as a client and/or as a mini-server. As a client, it requests media files from the system. A peer may opt to store segments of the media files that it has already consumed for a specific period of time. As a mini-server, it can provide these segments to other requesting peers in the system. We emphasize the miniature attribute of the mini-server, since the peer was never intended to function as a full server. Instead, it serves a few peers for a limited duration. Each of these mini-servers adds only a little to the overall system capacity. Combining a large number of them can significantly amplify the capacity of the system. Peers join the system along with their resources. More cooperating peers results in an increase in the system capacity. This leads to a scalable system that can potentially support an enormous number of clients.

The system as a whole benefits from the cooperative peers. A well designed peer-to-peer system should provide sufficient *incentives* to motivate peers to share their storage capacity as well as their network bandwidth. In a recent study of two popular peer-to-peer file sharing systems (Napster and Gnutella), Saroui *et al.* [11] discovered that peers tend to avoid sharing their resources without enough incentives. The incentives may include lower rates (\$/Byte) for those who store and supply media files to other peers in the system. Another way to encourage peers to share their resources is the "rewards for sharing" mechanism [5]. By this mechanism, points or credits are given to a cooperative peer as it increases the sharing. Consuming peers, get penalized by paying more to get resources from the system. In [6],

we propose and analyze a simple *revenue sharing* incentive mechanism by which a service provider can motivate peers to contribute resources to the system.

2.1. The Model

The P2P model consists of a set of peers. We have a set of seeding peers that provide or *seed* the newly published media files into the system. They stream these files to a limited number of peers, which in turn, will feed another set of peers. After a short period of time, the system will have sufficient peers that already have the newly published media to satisfy almost all requests for the file without having to overload the seeding peers. We formally define the entities involved in our model as well as their roles and how they interact with each other in the following.

- Peers. This is the set of nodes currently participating in the system. We denote P = {P₁, P₂, ···, P_N} as the set of all peers in the system. Every peer P_i, 1 ≤ i ≤ N, specifies three parameters: (1) R_i (in Kb/s), the maximum rate peer P_i is willing to share with others; (2) G_i (in bytes), the maximum storage space the peer is willing to allocate to store segments of one or more media files; and (3) C_i, the maximum number of concurrent connections that can be opened to serve requesting peers. By using these three parameters, a peer has the ability to control its level of cooperation with other peers in the system.
- Seeding peer. One of the peers or a subset of them
 may seed the new files into the system. We chose the
 name seeding peers to indicate that their main functionality is to initiate the service and not to serve all
 clients at all times.
- 3. **Stream.** A stream is a time-ordered sequence of packets belonging to a specific media file. This sequence of packets is not necessarily downloaded from the same serving node. Neither is it required to be downloaded in order. It must be displayed by the client in a specific order. It is the responsibility of the *scheduler* to download the packets from a set of possible nodes before their scheduled display time to guarantee non disruptive playing of the media.
- 4. **Media files.** The set of movies currently available in the system. Every movie has a size in bytes, and is recorded at a specific bit rate *R* Kb/s. A media file is divided into *N* segments. A segment is the minimum unit which a peer can cache. A supplying peer may provide the cached copy of the segment at a rate lower than the required rate *R*. In general, one segment can be streamed to the requesting peer from multiple peers



at the same time. According to our protocol (see Section 3), every peer will supply a different *piece* of the segment proportional to its streaming rate.

3. P2P Streaming Protocol

In this section, we describe the building blocks of the protocol used by a participating peer in the system. The protocol is composed of three phases and is to be run by a peer requesting a media file (pseudo code is give in [6]). In phase I, the requesting peer checks for the availability of the desired media file in the system. The phase starts with a crucial *searching* step. We describe the searching technique in Section 4.

The information returned by the searching step is arranged into a two-dimensional table. Each row j of the table contains peers that are currently caching segment s_i of the requested file. Certain information about each peer is stored; e.g., its IP address, the available streaming rate, and some reliability information from the peer's history. Each row is then sorted to select the most suitable peers to stream from. Several criteria can be used for sorting, such as proximity to the client (in terms of network hops), available streaming rate, and peer's average on-line time. A weighted sum of some (or all) criteria could also be used. In our experiments, we use the proximity as the sorting criterion. This reduces the load on the network, since traffic will traverse fewer domains. In addition, the delay is expected to be shorter and less variable, i.e., smaller jitter. Phase I ends with a verification step to make sure that all segments are available. Otherwise, the requesting client backs off and tries later after exponentially increasing the waiting time.

The streaming phase starts only if phase I successfully finds all segments. Phase II streams segment by segment. It overlaps the streaming of one segment with the consumption of the previous segment. The playback of the media file starts right after getting the first segment. Because of the variability in network and peer conditions, buffering few segments ahead would result in a better playback of the media file. The buffering time can hide transient extra delays in packet arrivals. In case that one of the supplying peers fails or goes off line, this buffering time may hide delays due to finding and connecting to another peer from the standby table.

For every segment s_j , the protocol concurrently connects to all peers that are scheduled to provide pieces of that segment. The connections remain alive for time δ , which is the time to stream the whole segment. Different non-overlapping pieces of the segment are brought from different peers and put together after they all arrive. The size of each piece is proportional to the rate of its supplying peer. Let us define \mathbb{P}^j as the set of peers supplying segment j. If a peer $P_x \in \mathbb{P}^j$ has a rate $R_x \leq R$, it will pro-

vide $|s_j|(R_x/R)$ bytes starting at wherever peer P_{x-1} ends. Since every peer supplies a different piece of the segment and $\sum_{x=1}^{|\mathbb{P}^j|} |s_j|(R_x/R) \geq |s_j|$, all pieces of the segment will be downloaded by the end of the δ period.

Finally, in phase III, the peer may be allowed to cache some segments. This depends on the dispersion algorithm used. We present dispersion algorithms in Section 4.

4. Architecture

Two approaches may be used to realize the P2P streaming service model. The first approach relies on having a special entity to maintain information about the currently participating peers. We call it the *index* approach. If the seeding entity is a set of servers owned by a provider, the index will typically be maintained by this set of servers. The details of this approach are in the following subsections. The second approach does not assign special roles to any peer. It needs to logically interconnect peers in the system, we call this the *overlay* approach. We are currently working out the details of this approach. Both approaches follow the P2P paradigm, in which peers help each other in providing the streaming service. The two approaches are different in handling the preparatory steps of the streaming phase. The most important of these steps are: locating peers with the required media file (searching), and quickly disseminating media files into the system (dispersion).

Before we present the index approach, we describe the *client clustering* idea, which is a key issue in the architecture. A cluster is defined as a logical grouping of clients that are topologically close to each other and likely to be within the same network domain [7]. It is highly beneficial for both the client and the network if a request can be fulfilled by peers within the same domain. For the network, it means that the traffic will travel fewer hops and hence will impose less load on the backbone links. The traffic delay will be shorter and less variable within the same domain, which is a desirable property for the streaming service.

We use a client clustering technique similar to the one proposed in [7]. The technique uses routing tables gathered from several core BGP routers. Client IP addresses that have the same longest prefix match with one of the routing table entries are assigned the same cluster ID. To illustrate the idea, consider five peers P_1 , P_2 , P_3 , P_4 , and P_5 , with IP addresses 128.10.3.60, 128.10.3.100, 128.10.7.22, 128.2.10.1 and 128.2.11.43, respectively. Suppose that among many entries in the routing tables, we have the following two entries: 128.10.0.0/16 and 128.2.0.0/16. The first three peers (all within Purdue University) share the same prefix of length 16 with the entry 128.10.0.0/16 (Purdue domain) and a prefix of length 12 with the entry 128.2.0.0/16 (CMU domain). Therefore, peers P_1 , P_2 , and P_3 will be grouped together in one cluster with ID



128.10.0.0/16. Similarly, peers P_4 and P_5 will be grouped together in another cluster with ID 128.2.0.0/16. Notice that, using the same idea, a finer clustering within the same domain is also possible. For instance, P_1 and P_2 may be grouped in a smaller cluster with ID 128.10.3.0/24. This clustering technique does not incur much overhead, since it is performed once when the peer first joins the system.

4.1. Index Approach

The index approach requires one (or a small subset) of the participants to maintain an *index* to all other peers in the system. The index can be maintained by the same machine seeding the media files, or by a separate machine. In any case, we call the maintainer of the index as the index server. This approach may be described as a *hybrid* scheme because the streaming process is peer-to-peer, while the searching and the dispersion processes are server-assisted. The main role of this special node is *not* to provide the streaming service, but to facilitate the searching and the dispersion processes. The load, in terms of CPU, bandwidth, and storage, imposed by the control information required by the searching and dispersion processes is a small fraction of the load imposed by the streaming service. To some extent, this alleviates the scalability and the single point of failure concerns that typically arise in such architectures. This approach greatly simplifies the searching process and reduces the overhead associated with it. Without the index, the overhead traffic puts a non-negligible load on the system. The index approach is practically easier and faster to deploy and more appropriate for a commercial media provider, since a commercial media provider would keep a server for accounting and charging customers and to seed the newly available media files into the system.

4.1.1 Index Searching

A key issue in the index approach is to keep the index current. First, notice that peers who are currently caching some of the media files are known to the index. Because they initially contact the index server to get served those media files. And, it is the index server that decides for them what to cache, as explained in the next subsection. Therefor, the index already knows who has what. The index server, though, does not know whether a peer is currently on or off line. Several techniques may be employed to keep the index up to date. In the case that a peer gracefully shuts down, a daemon running on the peer can send a notification message to the index server. Since it is unlikely that too many peers shut down synchronously, these notification messages will not cause message implosion at the index server. Another way to keep the index server current is to have the requesting client checks the list of candidate peers returned by the

Algorithm IndexSearch

```
/* Index server: upon receiving a query from peer P_r */ c \leftarrow getCluster(P_r) for j=1 to N do /* for every segment in the file */ candList[j] \leftarrow peers in c that have segment s_j if \sum_{P_x \in candList[j]} R_x < R then if Peers from other clusters can provide the shortage then Append to candList[j] sufficient peers from the closest clusters else return empty list to P_r /* P_r backs off */ end if end for return candList to P_r
```

Figure 1. Index-based Searching algorithm

index server by, for example, pinging them. The client then reports to the index server the status of all peers in the candidate list in one message.

Figure 1 summarizes the searching process in the index approach. We assume that the index server gets the BGP routing tables and builds the clustering database apriori. Upon receiving a query from a client asking for a specific file, the index server first identifies the cluster to which the client belongs. If peers within the same cluster can satisfy the request, those peers will be returned to the client as a set of candidates to stream the request. Otherwise, peers from the closest clusters are chosen to serve the request. To find the closest clusters in terms of network hops, the same clustering idea can be applied recursively, that is, several smaller clusters are grouped together into a larger cluster if they share the same common network prefix. The index server, then, tries to satisfy the client's request from the larger cluster. For example, if we have peers P_1, P_2, P_3, P_4 , and P_5 , as described above, and P_1 is requesting a file. The index server will first try to satisfy the request from peers located within the cluster with ID 128.10.3.0/24, i.e., from peer P_2 . If P_2 can not fulfill the request, the index server will try peers within the larger cluster with ID 128.10.0.0/16, i.e., from peers P_2 and P_3 . If P_2 and P_3 can not fulfill the request, the index server will try to find peers from other clusters to make up the shortage. If the request can be fulfilled by any set of peers, this set is returned to the requesting client as a list of candidate peers. If the system does not have sufficient capacity, an empty candidate peers list is sent to the client. The client then backs off and tries after an exponentially increased waiting time.

4.1.2 Index Dispersion

Caching the right segments of the media file at the right places is crucial to the incremental expansion of the system capacity. The objective of the dispersion algorithm is to store enough copies of the media files in each cluster to serve all expected client requests from that cluster. As de-



scribed in [6], peers are given *incentives* to cooperate; especially, if the service is provided by a commercial provider. These incentives are costs imposed on the provider. For this reason, it is important to keep just the required capacity in the system. To do so, we propose a dynamic dispersion algorithm that adjusts the capacity within each cluster according to the average number of client requests from that cluster.

The dispersion algorithm works in the following setting. At a specific instant of time, the system can serve a certain number of requests concurrently. A client P_y sends a request to the system to get the media file. The client also declares its willingness to cache up to N_y segments to serve them to other clients with rate R_y in the future. The dispersion algorithm decides whether or not this peer should cache, and if so, which specific segments it should cache. The algorithm should ensure that, on the average, the same number of copies of each segment is cached, since all segments are equally important. To clarify, consider a file with only two segments. Keeping 90 copies of segment 1 and 10 copies of segment 2 means that we have effectively 10 copies of the media file available. In contrast, keeping 50 copies of each segment would result in 50 copies of the media file.

The IndexDisperse algorithm, shown in Figure 2, is to be run by the index server. Consider one media file with N segments, rate R Kb/s, and duration T hours. The algorithm requires the index server to maintain three types of information: per-peer information, per-cluster information, and per-system (or global) information.

For every peer P_x , the index server maintains: (1) N_x , the number of segments which are currently cached by P_x ; (2) R_x , the rate at which P_x is willing to stream the cached segments; and (3) u_x , $0 \le u_x \le 1$, the fraction of time P_x is online. Recall that the peer is not available all the time.

For every cluster c, the index server maintains the following: (1) $L_{\rm c}$, $1 \le L_{\rm c} \le N$, the next segment to cache. (2) $q_{\rm c}$, the average request rate (per hour) the media file is being requested by clients from c. $q_{\rm c}$ represents the required capacity in the cluster c per hour. (3) $a_{\rm c}$, the average number of copies of the movie cached by peers in cluster c. c is computed from the following equation:

$$a_{c} = \sum_{P_{x} \text{ in } c} \frac{R_{x}}{R} \frac{N_{x}}{N} u_{x}. \tag{1}$$

The summation in Equation (1) computes the effective number of copies available in the cluster. It accounts for two facts: first, peers are not always online (through the term u_x), and second, peers do not cache all segments at the full rate (through the term $R_x N_x/RN$). Dividing $a_{\rm c}$ by T results in the number of requests that can be satisfied per hour, since every request takes T hours to stream. Hence,

Algorithm IndexDisperse

```
L_c \leftarrow 1, \forall \mathsf{c} while TRUE do Wait for a caching request /* Got request from peer P_y to cache N_y segments with rate R_y */ \mathsf{c} \leftarrow getCluster(P_y) /* identify client's cluster */ Compute a_c, q_c, A, Q if q_c > a_c or Q \gg (1/T)A then /* need to cache in round robin */ if (L_c + N_y - 1) \leq N then L_e = L_c + N_y - 1 else L_e = N_y - (N - L_c + 1) end if Peer P_y caches from segment L_c to segment L_e to d if end while
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Figure 2. Index-based dispersion algorithm.

 $(1/T)a_{\rm c}$ represents the available capacity in the cluster c per hour.

The index server maintains two global variables: (1) $A = \sum_{c} a_{c}$, the average number of copies of the movie cached by all peers in the system. (2) $Q = \sum_{c} q_{c}$, the average movie request rate in the system. Q and (1/T)A represent the global required capacity and the global available capacity in the system, respectively.

The algorithm proceeds as follows. Upon getting a request from peer P_y to cache N_y segments, the index server identifies the cluster c of the requesting peer. Then, it computes a_c , q_c , A, and Q. The algorithm decides whether P_y caches based on the available and the required capacities in the cluster. If the demand is larger than the available capacity in the cluster, P_y is allowed to cache N_y segments in a cluster-wide round robin fashion. To clarify, suppose we have a 10-segment file. L_c is initially set to 1. If peer P_1 sends a request to cache 4 segments, it will cache segments 1, 2, 3, and 4. L_c , the next segment to cache, is now set to 5. Then, peer P_2 sends a request to cache 7 segments. P_2 will cache segments 5, 6, 7, 8, 9, 10, and 1. L_c is updated to 2, and so on. This ensures that we do not over cache some segments and ignore others.

Furthermore, the IndexDisperse algorithm accounts for the case in which some clusters receive low request rates while others receive very high request rates in a short period. In this case, the global required capacity Q is likely to be much higher than the global available capacity (1/T)A, i.e., $Q\gg (1/T)A$. Therefore, even if the intra-cluster capacity is sufficient to serve all requests within the cluster, the peer is allowed to cache if $Q\gg (1/T)A$ in order to reduce the global shortage in the capacity. The operator \gg used in comparison is relative and can be tuned experimentally.



5. Evaluation

We study the performance of the P2P model under various situations, e.g., different client arrival patterns and different levels of cooperation offered by peers. We simulate a large (more than 13,000 nodes) hierarchical, Internet-like, topology. We use the GT-ITM tool [1] for generating the topology and the Network Simulator *ns-2* [8] in the simulation. Due to space limitations, we present a sample of the results. The reader is referred to the technical report [6] for a detailed description of the simulation as well as more results including results for Poisson and flash crowd arrivals, and the evaluation of the dispersion algorithm.

We simulate the following scenario. A seeding peer with a limited capacity introduces a media file into the system. According to the simulated arrival pattern, a peer joins the system and requests the media file. Then, the P2PStream protocol, described in Section 3, is applied. We do not assess the overhead imposed by the searching step in this set of experiments. If the request can be satisfied, i.e., there is a sufficient capacity in the system, connections are established between the supplying peers and the requesting peer. Then, a streaming session begins. The connections are over UDP and carries CBR traffic. If the requesting peer does not find all segments with the full rate, it backs off and tries again after an exponentially increased waiting time. If the waiting time reaches a specific threshold, the request is considered "rejected" and the peer does not try again. When the streaming session is over, the requesting peer caches some of the segments depending on the level of cooperation, called the caching percentage. For instance, if the caching percentage is 10% and the media file has 20 segments, the peer stores two randomly-chosen segments. The peer also selects a rate at which it wants to stream the cached segments to other peers.

Figure 3 shows how the system capacity evolves over the time. The average service rate increases with the time, because as the time passes more peers join the system and contribute resources to serve other requesting peers. The average waiting time, shown in Figure 4, is decreasing over the time, even though the system has more concurrent clients. This is due to the rapid capacity amplification. The capacity is rapidly amplified, especially with high caching percentage. For instance, with 50% caching, the system is able to satisfy all the requests submitted at 5 requests/minute after about 250 minutes from the starting point.

Figure 5 verifies the diminishing role of the seeding peer. Although the number of *simultaneous* clients increases until it reaches the maximum (limited by the arrival rate), the proportion of these clients that are served by the seeding peer

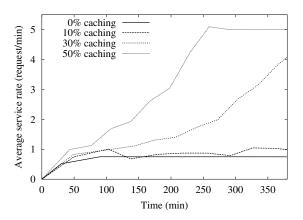


Figure 3. Average service rate.

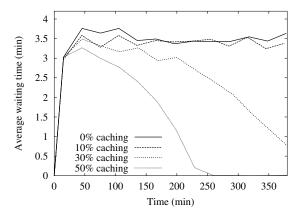


Figure 4. Average waiting time.

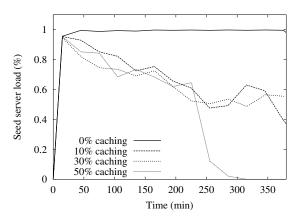


Figure 5. Load on the seeding peer.



decreases over the time, especially with high caching percentages. For instance, with 50% caching and after about 5 hours, we have 100 concurrent clients (6.7 times the original capacity) and none of them is served by the seeding peer. Reducing the load on the seeding peers is an important feature of the P2P streaming architecture, because it means that the seeding peers need not to be powerful machines with high network connectivity. Besides being moderate machines, the seeding peers are used only for a short period of time. Therefore, the cost of deploying and running these seeding peers (in case of a commercial service) is greatly reduced.

6. Related Work

Significant research effort has addressed the problem of efficiently streaming multimedia, both live and on demand, over the best-effort Internet. Directly related to our work are systems like *SpreadIt* [2] for streaming live media and *CoopNet* [10], [9] for both live and on-demand streaming. Both systems build distribution trees using application-layer multicast and, like ours, they rely on cooperating peers. Multicast (network- or application-layer) is the basis for several other media delivery systems [3] [4]. Our work is different from these systems, since we do not use multicast in any form and our system is more appropriate for ondemand media service.

In the client/server world, proxies and caches are deployed at strategic locations in the Internet to reduce and balance load on servers and to achieve a better service. Content Delivery Network (CDN) companies such Akamai and Digital Island follow similar approaches to provide media streaming and other services. Our approach does not require any powerful proxies or caches. Rather, it uses peers' extra resources as numerous tiny caches. These tiny caches do not require large investment and collectively enlarge the capacity of the system in a way that potentially outperforms any powerful caches.

7. Conclusions and Future Work

We presented a P2P media streaming model that can serve many clients in a cost effective manner. We presented the details of the model and showed how it can be deployed over the current Internet. Specifically, we presented a P2P streaming protocol used by a participating peer to request a media file from the system; a cluster-based dispersion algorithm, which efficiently disseminates the media into the system; and a searching algorithm to locate nearby peers who have segments of the requested media file. Through a large-scale simulation, we showed that our model can handle several types of client arrival patterns, including suddenly increased arrivals, i.e., flash crowds.

We are currently embarking on implementing a prototype of the P2P media streaming system. The objective is to better assess to the proposed model and to demonstrate its applicability for a wide deployment. Addressing the security and robustness issues of the model are parts of our future work.

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