

*Communication Networks*Proxy caching algorithms and implementation
for time-shifted TV servicesTim Wauters^{1*}, Wim Van De Meerssche¹, Peter Backx¹, Filip De Turck¹, Bart Dhoedt¹,
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SUMMARY

The increasing popularity of multimedia streaming applications introduces new challenges in content distribution networks (CDNs). Streaming services such as Video on Demand (VoD) or digital television over the Internet (IPTV) are very bandwidth-intensive and cannot tolerate the high start-up delays and poor loss properties of today's Internet. To solve these problems, caching (the initial segment of) popular streams at proxies could be envisaged. This paper presents a novel caching algorithm and architecture for time-shifted television (tsTV) and its implementation, using the IETF's Real-Time Streaming Protocol (RTSP). The algorithm uses sliding caching windows with sizes depending on content popularity and/or distance metrics. The caches can work in stand-alone mode as well as in co-operative mode. This paper shows that the network load can already be reduced considerably using small diskless caches, especially when using co-operative caching. A prototype implementation is detailed and evaluated through performance measurements. Copyright © 2007 John Wiley & Sons, Ltd.

1. INTRODUCTION

During the last few years, the architectural model of access networks has evolved towards multi-service and multi-provider networks. Ethernet as well as full IP alternatives have been investigated as viable connectionless successors, for the legacy ATM-based platforms. While the introduction of Ethernet up to the edge (e.g. through VLANs) solves some of the existing PPP problems, new ones are created. Traffic segregation issues, because of address resolution complications, prevent large-scale access network deployments and, therefore, an IP-aware network model [31] is often considered a valuable alternative.

One of the emerging services is television over IP. Currently, IPTV services are generally limited to Video on Demand (VoD, per TV program) and Broadcast TV (per TV channel). Video servers are typically located at the edge

of the core network and put a heavy burden on the access networks in case of large deployments, possibly causing the network to congest. The solution proposed in this paper is to focus on time-shifted television (tsTV) and to deploy additional distributed caches and streamers in the aggregation nodes, co-operating on both a peer-to-peer and a hierarchical level. This approach offers an alternative for deploying dedicated home equipment for video storage, such as a home Personal Video Recorder (PVR), that has limited throughput capacity and is rather expensive. Time-shifted TV enables the end-user to watch a broadcasted TV program with a time shift, that is the end-user can start watching the TV program from the beginning, although the broadcasting of that program has already started or is already finished.

As shown in Figure 1, the popularity of a television program typically reaches its peak value within several minutes after the initial broadcast of the program and

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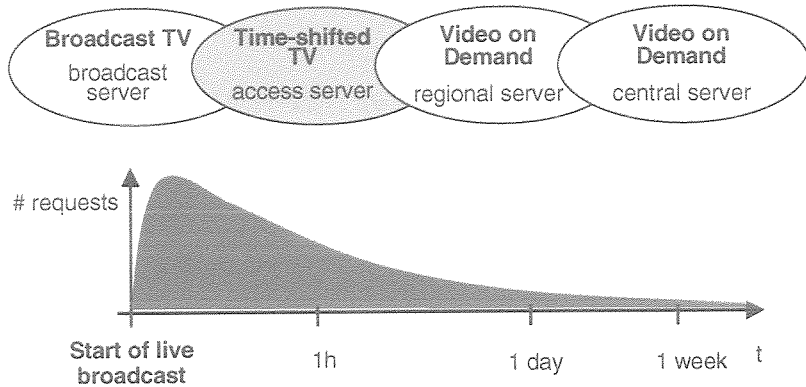


Figure 1. Delivery mechanisms for IPTV.

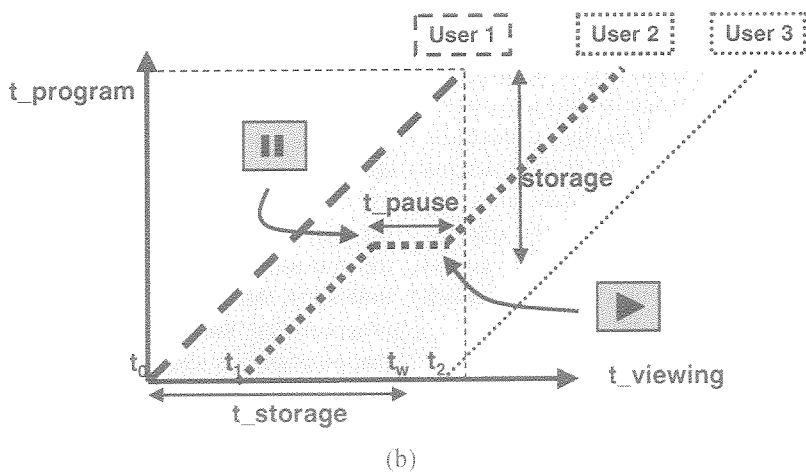
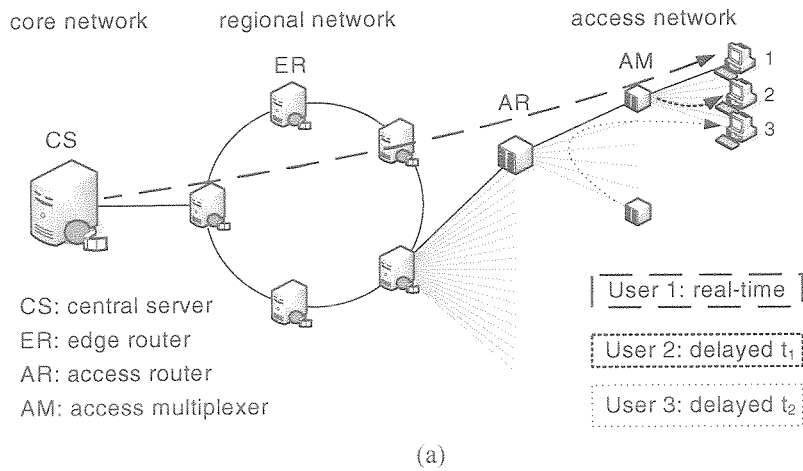


Figure 2. Time-shifted television: (a) typical network topology and (b) tsTV streaming diagram.

exponentially decreases afterwards. This means that caching a segment with a sliding window of several minutes, for each current program can serve a considerable part of all user requests for that program from start to finish.

In Figure 2a, user 1 is the first to request a certain television program and gets served from the central server. Afterwards, other requesting users (e.g. user 2) can be served by the proxy, as long as the window of the requested program is still growing. After several minutes, the window stops growing and begins sliding, so that user 3 cannot be served anymore and will be redirected to the (central or regional) server or, in case of co-operative caching, to a neighbour proxy with the appropriate segment, if present. Another view of the same situation is given in Figure 2b. User 1 watches the program in real time, user 2 can be served within the cached window, while user 3 cannot. Pausing (parallel to the horizontal axis) can also be supported within the segment window, as well as fast forward or rewind (parallel to the vertical axis).

The above-mentioned servers can be distinguished as follows, based on their location and the streaming service offered:

- The *central server* stores all available content (TV programs), but typically only has to serve requests for less popular or older programs.
- The *regional server* generally stores more popular programs, such as recently broadcasted series or talk shows. These servers are located at the edge of the core network (see Figure 2a), thereby restricting the tsTV video traffic within the access networks.
- The *proxy servers* are located close to the users and only store fragments of the most popular content. Contrary to the traditional VoD technology service that would best fit the central and regional servers, the optimal technology choice for these proxy servers is to make use of a caching strategy, dynamically tailored to the tsTV service consumption pattern. Furthermore, interactive commands can be handled by these proxy servers, as long as the time-shifts remain within the stored segment window.

The remainder of this paper is structured as follows. Research work related to this research is briefly discussed in Section 2. Section 3 presents an analytical model of the sliding-interval caching problem with fixed window sizes, for comparison with our caching algorithms and to have an initial estimate of the required storage space. The next section introduces our sliding-interval caching algorithm, for both stand-alone and co-operative caching. It determines the location and the size of the different segments at the proxy caches, at run-time. In Section 5, the Real-Time

Streaming Protocol (RTSP) implementation is studied more detail and evaluated through measurements. Section 6 concludes this paper and presents ideas for future work.

2. BACKGROUND AND RELATED WORK

In this paper, a greedy sliding-interval co-operative caching algorithm will be presented. This section gives an overview of existing solutions in research and explains this choice.

Previous studies on proxy caching techniques [2] or distributed replica placement strategies for content distribution networks (CDNs) [3–8] show that greedy algorithms that take distance metrics and content popularity into account perform better than more straightforward heuristics, such as least recently used (LRU) or least frequently used (LFU).

Segment-based caching techniques have been studied extensively for streaming media, due to the huge size of multimedia streams, compared to traditional web objects. A survey on different strategies such as prefix caching [9], segment caching [1, 10–12], rate-split caching [13] and sliding-interval caching [14, 15] has been presented in Reference [2]. The main goal of prefix caching is to reduce the start-up delay by caching the initial portion of the stream at the proxy. This paradigm is generalised to segment caching, where cache decisions are made for a series of segments of the stream. In rate-split caching, time partitioning is done along the rate axis, instead of along the time axis. This way, the cache takes care of the peak rate in VBR streaming, while the backbone only has to cope with the lower constant rate.

Of particular interest for this study is sliding-interval caching, where the cached portion of the stream is initially a growing prefix, but afterwards, a dynamically updated sliding interval. This way, consecutive requests can be served from start to finish within this window. Since most tsTV service requests are expected to arrive early, after the start of the initial broadcast of a prime-time program, a window size of several minutes can be sufficient. Another advantage of this technique is the support of interactive operations such as pause, fast-forward and rewind, at least within the segment interval.

A more advanced aspect is the use of co-operative proxy caching [16–18], where a better performance than with independent proxies can be achieved through load balancing and improved system scalability. In this case, it is important to continuously keep track of cache states. Note that contrary to standard co-operative proxy caching, there is no need to switch to segments on other proxies, when using co-operative proxy caching with sliding interval

Similar peer-to-peer caching techniques have also been introduced in streaming CDNs, where whole files are stored instead of segments [19].

Several studies such as Reference [20] have been investigating the implementation of segment-based caching techniques on proxies using the RTP/RTCP/RTSP protocol suite.

The originality of this work is in the combination of the abovementioned techniques, applying the p2p and caching mechanisms from previously studied VoD content placement algorithms to sliding-interval caching. The proposed storage model is evaluated and implemented for IPTV, as a novel time-shifted TV service. The RTSP protocol allows for transparent request forwarding, which further optimises the content placement by creating one large virtual cache.

3. ANALYTICAL APPROACH

Before presenting our sliding-interval caching algorithm, we introduce an analytical model of a tsTV solution based on sliding-interval caching with fixed window sizes, offering a method to estimate the required storage space in the network.

3.1. Model parameters

Consider a model where each TV program is characterised by a start time τ_i , a duration T_i and a function $\lambda_i(t)$, representing the request arrival rate for this program. $N(t)$ denotes the total number of programs with $\tau_i \leq t$. The proxy cache I , placed between the server and the clients, contains the first X min of any currently streaming file with $t - T_i \leq \tau_i \leq t$.

3.2. Cache hit rate

We derive an expression for the hit rate of cache I , $h_I(t)$. Consider further the time period $[t, t + \Delta t]$, then the total number of requests is given by

$$\sum_{i=1}^{N(t)} \lambda_i(t) \Delta(t).$$

To find the total number of successful requests (i.e. requests that can be served by the cache) for the currently broadcasted program j in a single channel situation, we assume a uniform distribution for τ_j and make the following observations:

- these requests have to arrive at most X minutes after τ_j
- only a fraction X/T_j of the requests is served from cache I

Therefore the total number of successful requests is given by

$$\lambda_j(t) \Delta(t) \frac{X}{T_j}$$

Averaging over all programs j for which $t - X \leq \tau_j \leq t$, multiplying by the total number of channels K and supposing that popularity and duration are uncorrelated, we obtain the following expression:

$$h_I(t) = K \frac{\langle \lambda_i(t) \rangle * \frac{X}{\langle T \rangle}}{\sum_{i=1}^{N(t)} \lambda_j(t)},$$

with $\langle \rangle *$ denoting averaging, on the condition that $t - X \leq \tau_j \leq t$. Supposing further that λ_i is a separable function of i and t , such that $\lambda_i(t) = \lambda_i f(t - \tau_i)$, with $f(t)$ a normalised function such that $f(t) = 0$ for $t < 0$ and

$$\int_0^{\infty} f(t) dt = 1,$$

we can write:

$$\begin{aligned} \langle \lambda_j(t) \rangle * &= \langle \lambda_j \rangle \langle f(t - \tau_j) \rangle * \\ &= \frac{\langle \lambda_j \rangle}{X} \int_0^{\infty} f(t) dt \end{aligned}$$

as long as $X < \langle T \rangle$. Hence,

$$h_I(t) = K \frac{\langle \lambda \rangle \int_0^X f(t) dt}{\langle T \rangle \sum_{i=1}^{N(t)} \lambda_i(t)}$$

Further consider a time period P , then the total number of broadcasted programs is $N(P) = KP / \langle T \rangle$. Suppose user group of size G , each requesting r programs per second on average, then the total number of requests is given by GrP . Therefore, the average number of requests for a long enough period of time will satisfy

$$\langle \lambda \rangle = \frac{GrP}{N(P)} = \frac{GrP}{KP / \langle T \rangle} = \frac{Gr \langle T \rangle}{K}$$

On the other hand, the total number of requests per time unit is given by

$$\sum_{i=1}^{N(t)} \lambda_i(t) = Gr,$$

simplifying our expression for the cache *I* hit ratio to

$$h_I = \int_0^X f(t)dt$$

Taking for *f(t)* an exponentially decreasing function $b \exp(-bt)$ (for $t > 0$), we get

$$h_I = 1 - e^{-bX}$$

as long as $X \ll \langle T \rangle$. The size of cache *I* is simply KX .

3.3. Example results

Figure 3 shows the server load for different values of the cached segment size. If the content popularity only decreases slowly (e.g. by 10% after each interval, $b = -\ln(0.9)/\Delta$), the server load cannot be reduced significantly. When the content popularity is halved after each interval ($b = -\ln(0.5)/\Delta$), the server load is halved as well when the segment size is Δ (Figure 3). It is then given by

$$1 - h_I = \left(\frac{1}{2}\right)^a,$$

if $X = a\Delta$.

Similar results for the server load can be found using the sliding-interval caching algorithm presented in the following section (compare the curve for $b = -\ln(0.5)/\Delta$ to the 's->c1' curve in Figure 10a, for stand-alone caches at level 2 and a halved content popularity after Δ).

4. SLIDING-INTERVAL CACHING ALGORITHM

Our caching algorithm for tsTV services is presented in this section. Since we assume that, in general, only

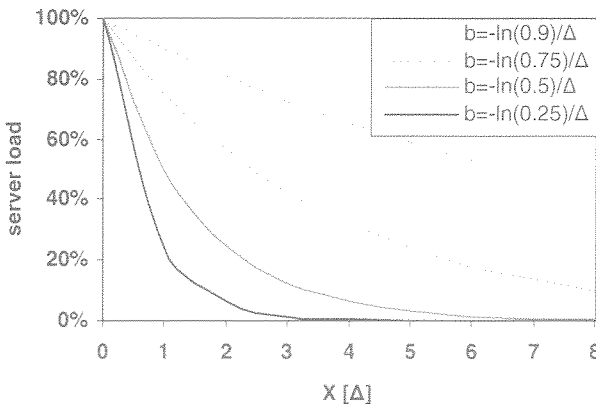


Figure 3. Analytical solution for the server load, for different values of the segment size.

segments of programs will be stored, cache sizes can be limited to a few gigabytes (corresponding to a few hours of streaming content). This way smaller streaming servers can be deployed closer to the users, without increasing the installation cost excessively.

4.1. Basic principle

The cache will be virtually split up in two parts: a small part *S* and a main part *L*. Part *S* will be used to cache the first few (e.g. 5) minutes of every newly requested (or broadcasted) program, mainly to determine its initial popularity. Its size is generally smaller than 1 GB (typically 1 h of streaming content).

Part *L* will be used to store the segments (with growing or sliding windows) of the currently most popular programs. The actual size of each segment in part *L* will be determined and, if necessary, adapted after each interval Δ (e.g. 5 min). During Δ , the cache is learning about the popularity of the programs.

Figure 4 shows the basic principle of the tsTV caching algorithm. During each interval Δ , program requests arrive at the different proxies. Each time, a parameter $A_{n,p}$ will be updated in proxy *n*, for program *p*. In general, this parameter tries to determine the popularity of the program, while taking distance metrics into account.

This means that a (segment of a) popular program might not be cached, because a nearby proxy already stores that (segment of the) program. $A_{n,p}$ is calculated as follows:

Every time a request for program p arrives at proxy n, $A_{n,p}$ is increased by 1 (popularity only) or by the hopcount between proxy n and the serving node (popularity and distance).

After each interval Δ , first all segments (sliding or growing) with status set to 'occupied' are stored in *L*. Afterwards, *L* is filled with segments with growing

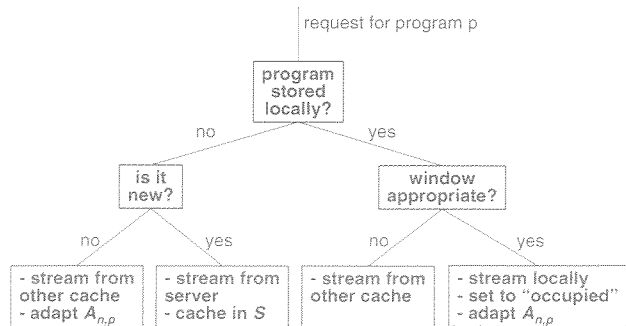


Figure 4. Basic principle of the tsTV caching algorithm at each proxy.

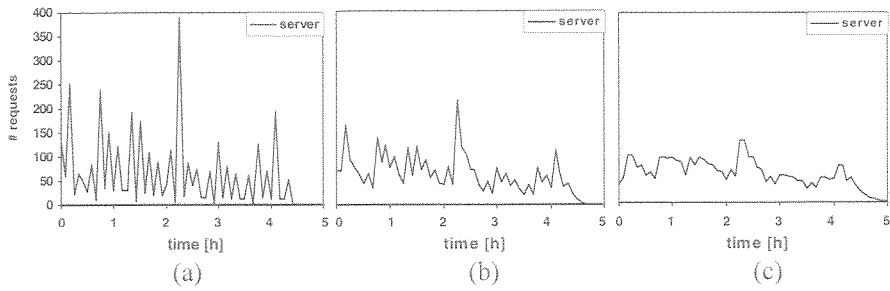


Figure 5. Server load without caches. All requests per program are made within 5(a), 30(b) or 60(c) min. 3000 requests are made total.

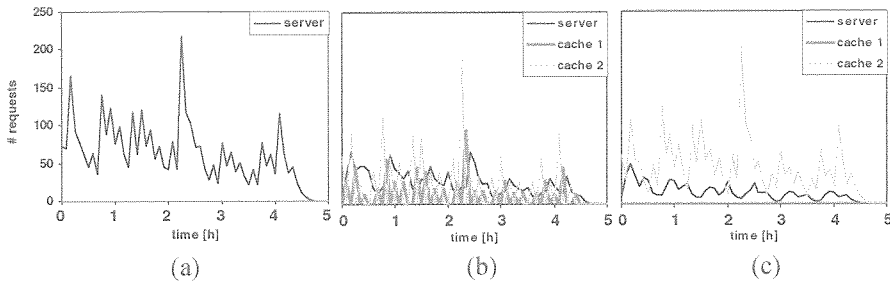


Figure 6. Server load with caches. All requests are made within 30 min. The cache sizes are 0 GB (a), 0.5 GB (b) and 4 GB (c).

windows for the most popular programs (i.e. with the highest values of $A_{n,p}$). All other segments are dropped, S is cleared and all values of $A_{n,p}$ are reset to 0.

The influence of premature termination, for example due to channel hopping, on the caching behaviour is much smaller than for the storage of whole video files [21], since the small cache part S , which is cleared after every learning interval Δ , handles most of these specific requests.

4.2. Numerical results for stand-alone caching

4.2.1. Input parameters

To illustrate the caching principle, a first set of simulations was performed on one branch of the access network tree of Figure 2a: a regional server with two hierarchical caches (Figure 7). The regional server offers 20 channels: 5 very popular channels (80% of all requests), 5 less popular channels (10% of all requests) and 10 unpopular channels (10% of all requests). The top five channels are served as a tsTV service, the other channels through standard VoD technology on the regional server.

The popularity of the programs per channel follows a Zipf-like distribution with parameter $\beta = 0.7$ (the popularity of the i 'th most popular program is proportional to $i^{-\beta}$). A total of 3000 requests are made during one

evening, of which 200, for the most popular program the most popular channel.

The popularity of a program reaches a peak during the first interval Δ ($=5$ min) and decreases exponentially afterwards (halved every interval Δ) (similar to Figure 5). Each channel offers six programs of 45 min per evening with a streaming bandwidth of 2.5 Mbps (1 GB per hour).

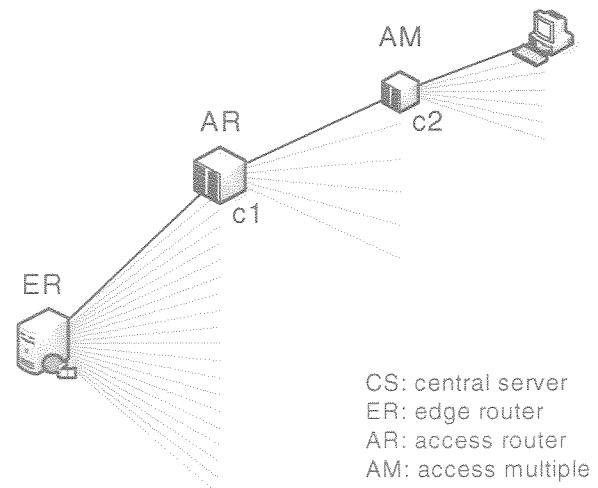


Figure 7. Basic access network topology.

Note that the values for these input parameters are very general, since Zipf-like content popularities are very commonly used in content distribution networks [22] and the relative results below are independent of the number of TV programs or their duration or bandwidth.

The caching algorithms have been implemented in C++ and evaluated in a standard discrete event simulator using the LEDA [23] library.

4.2.2. Server and cache load

In case no caches are in use, the load on the regional server is shown in Figure 5 (cumulated per interval Δ). The longer the period during which all requests are made, the smoother the traffic at the server (the total number of requests and the exponential decrease remain the same, while the initial popularity is different).

In Figure 6, caches are introduced. When both cache sizes are limited to 0.5 GB (S only: the number of channels times Δ or 25 min), the server load is already much lower and the caches serve most of the tsTV requests. What happens is that cache 1 (closest to the server) and cache 2 first store all 5-min prefixes of each new program, but since only cache 2 receives new requests afterwards, cache 1 will drop these segments after Δ . Afterwards cache 1 will store the next 5 min of each program, while cache 2 is storing the sliding ‘occupied’ windows from the first interval. This means that the caches serve all requests made during the first 10 min of each single program.

For infinite cache sizes (or 4 GB or higher in this example), the regional server only serves the VoD requests for channels 6–20. Cache 2 stores and serves all currently broadcasted programs.

More detail on the regional server and cache load is given in Figure 8 (tsTV only, top five channels). Note that the server load never drops to 0, since at least the first request for a certain program has to be served from the regional server. In Figure 9, the server load is shown for different values of the maximum request period per program. Since no upstream links are used in these simulations, the bandwidth on the links can easily be determined from the server and cache load.

4.3. Numerical results for co-operative caching

The same caching principles can be applied for a co-operative caching mechanism, where caches on the same level of the broadcast tree can collaborate, using peer-to-peer protocols to exchange information on stored content. Contrary to stand-alone caching, where a request that cannot be served is forwarded to the next cache on the path to

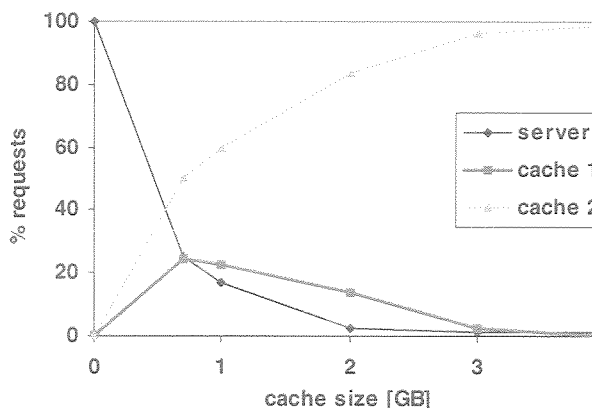


Figure 8. Server and cache load. All requests are made with 30 min.

the central server (hierarchical caching), caches can not forward requests to caches on the same level. However, the decision on when to store a certain fragment not only depends on the value of $A_{n,p}$, but also on the source not serving the request. Two different approaches have been implemented.

The first heuristic only takes the values of $A_{n,p}$ into account (‘cache from all sources’, CfA). This means that multiple caches store the same fragments, since content popularity is similar for most nodes. The numerical results will therefore be comparable to the results for stand-alone caching.

The second heuristic also takes the values for $A_{n,p}$ into account, but never stores content that is already stored at another cache (‘cache from server only’, CfS). This way the central server will be offloaded considerably, even with small caches, but many requests will have to be served by other caches over the access network links.

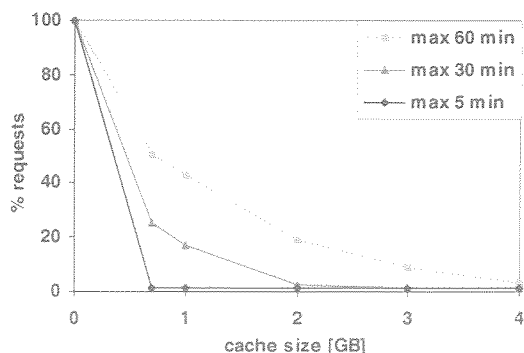


Figure 9. Server load for different values of the maximum request period.

Both alternatives have their benefits (the first one is optimal in case of larger caches, the second one in case of small caches). The optimal heuristic, however, takes the best of both worlds, storing unique content segments in one part of cache L (called L_1 , used in the CfS heuristic) and locally popular segments in another part (called L_2 , used in the CfE heuristic).

This way, the central server load is always minimised first: the expected server load using the storage space combining all parts L_1 can then be determined out of Figure 3. The access network load can be reduced afterwards, if the cache space is large enough. This heuristic is called 'cache from elected sources' (CfE).

4.3.1. Input parameters

The input parameters for the simulations are the same as in the previous section. The network topology (similar to Figure 7) now consists of a central server, one node at level 1 (without storage capabilities) and six proxy caches at level 2. The level 1 node is connected to the level 2 caches with bidirectional links, so that cache co-operation is possible.

Note that no storage space is available at the level one node so that the results of the simulations for cache co-operation are not influenced by hierarchical caching.

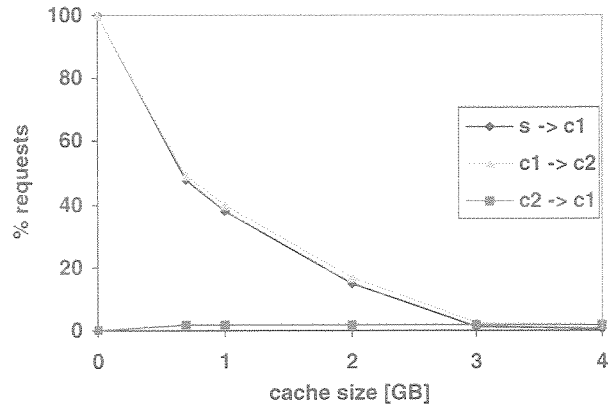
The cost of using the link from the central server to the node at level 1 has been set to 10 (in fact, any value higher than 1 will do), instead of 1. This way, the central server will be avoided when the requested segment can already be found on a neighbour level 2 cache (when calculating the shortest path, using the weighted Dijkstra algorithm).

4.3.2. Server, cache and network load

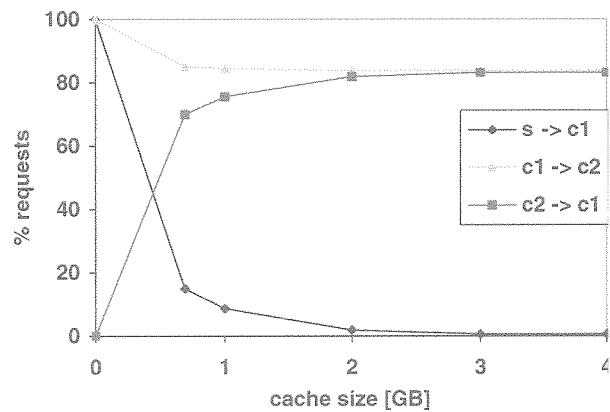
In case of stand-alone caching, the network bandwidth can easily be determined out of the cache and server load (Figure 8), since only downstream traffic is present on the access network. With co-operative caching, the uplinks in the access network are used as well.

Using the CfA heuristic (Figure 10a), the server load is almost identical to the case where stand-alone caches on level 2 are used. The only difference is that the central server does not need to serve the first stream to all of the six proxies, but only to one of them. Again the central server load for the tsTV channels drops to (almost) zero when 4 GB caches would be used. The uplinks from the level 2 caches to node 1 are almost never used, since all caches store the same fragments. The results are, therefore, very similar as for stand-alone caching (remember the analytical results of Figure 3, with 1 GB = 10 min per channel = 2Δ).

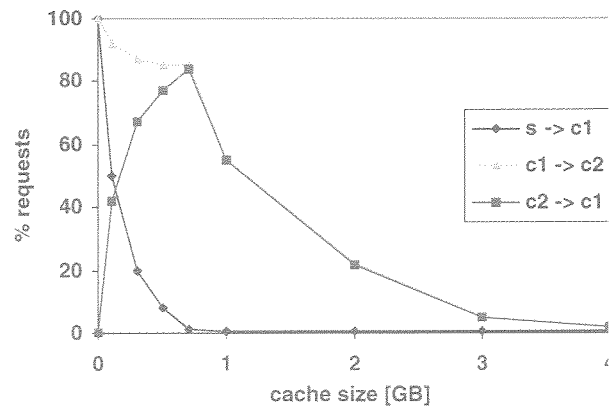
When the CfS heuristic is used (Figure 10b), each 5-min (Δ) fragment is only stored on one cache. This way,



(a)



(b)



(c)

Figure 10. Fraction of the streams on the links between the server and the level 1 node ($s \rightarrow c1$) and between the level 1 and two nodes (downlink $c1 \rightarrow c2$ and uplink $c2 \rightarrow c1$) for the CfA (a), CfS (b) and CfE (c) heuristics.

the central server load is already almost zero for the tsTV channels, when only 0.5 GB caches are used. The total storage space is then 3 GB, therefore, one could expect that the results for the central server load would correspond to the situation with 3 GB caches in stand-alone mode. This is not entirely the case, since it is possible that the first requests for a new program arrive at caches that have no storage place left in L_1 . These first requests are then served by the central server. The ‘core network load’ (represented by the link ‘s -> c1’) is reduced considerably, while the ‘access network load’ (represented by the links ‘c1 <-> c2’) is load balanced.

The *CfE* heuristic (Figure 10c) offers the best of both worlds. The server load is reduced effectively, while, in case of larger caches, the access network is offloaded as well. The server load (link ‘s -> c1’) is even lower then for the *CfS* heuristic. This is due to the RTSP request forwarding mechanism, allowing requests that arrive at a cache that has no storage space left in L_1 , to be forwarded automatically to another cache with enough storage space. This way the virtual cache consisting of all parts L_1 is filled up in an optimal way.

5. PROXY IMPLEMENTATION

A transparent RTSP proxy for time-shifted TV has been implemented (in C++) for evaluation purposes. This section gives an overview of the different components and protocols used and evaluates a prototype through performance measurements.

5.1. Functionality

In order to implement the proxy, its functionality is divided into logical parts. The communication with the users and the central server includes messages containing data about which program or channel has to be streamed, or VCR like commands such as PAUSE and STOP. A protocol, commonly used for this interaction is RTSP [24]. The streams themselves are encapsulated and delivered with Real-Time Protocol (RTP), a standard protocol for live streamed media [25].

A first functional component of the proxy is the *RTSP Proxy*, a component that communicates with the tsTV clients and the server using RTSP, interprets their messages and commands the other components to execute these requests. The *RTSP Proxy* component delegates the caching algorithm decisions to another component, the *Cache Verdict Manager*, a component that uses information from the *Cache State Manager*, which is updated through a centralised or distributed Cache State Exchange (CSE)

protocol. The task of the *Cacher* component is to store popular streams, sent to the proxy by the server (or another cache), in sliding windows. The streams are sent to the clients from these windows, a function that is handled by the *Streamer* component. The proxy also keeps track of the streams that are being sent to the proxy (which program channel, starting time, ...), through the *Stream Tracker* component, with help from the *Program Guide* component, which communicates with the electronic program guide (EPG) server. The *Packet Handler* acts as an interface dealing with low-level network interaction. Figure 11 gives an overview of the different components.

5.2. Detailed scenario

Figure 12 shows a detailed setup of a streaming session between the client, the proxy caches and the server.

First, the client sends an RTSP request to the server, but this request is intercepted by the proxy. In a first scenario (part 1a in Figure 12), the proxy does not store the requested fragment, forwards the request (with the destination address of the proxy) to the server, starts caching the stream from the server and forwards the RTP stream to the user. Afterwards, the proxy exchanges its new cache state in a distributed way to all other caches (part 2a in Figure 12). In a second possible scenario (part 1b in Figure 12), the proxy does not store the requested fragment and decides not to forward the request to another (proxy) cache, keeping the destination IP address of the client.

The other proxy decides to forward the request to the server, caches the fragment locally and sends the RTP stream directly to the client. Afterwards, the new cache states are exchanged through a centralised CSE protocol.

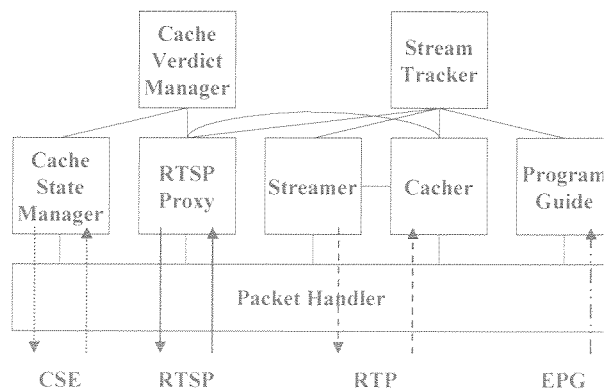


Figure 11. Overview of the different components in the proxy cache.

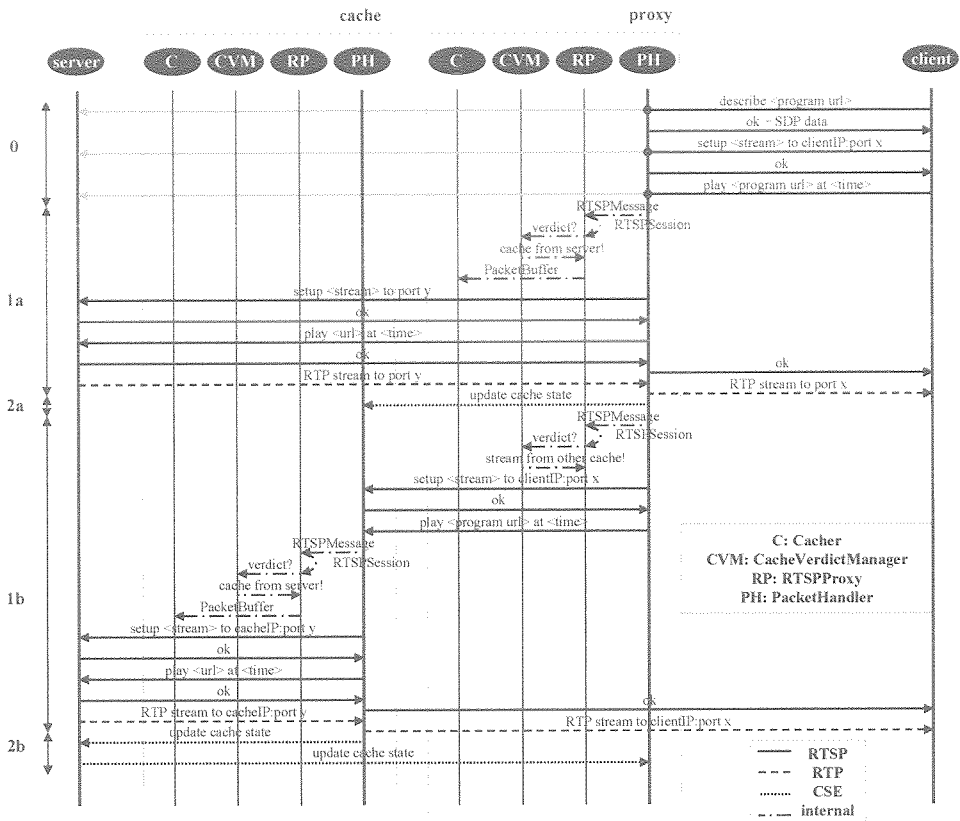


Figure 12. Detailed setup of a streaming session between client, proxy, any other cache and the server. In scenario a, the proxy caches the requested program from the server; in scenario b, the proxy forwards the RTSP request transparently to another cache.

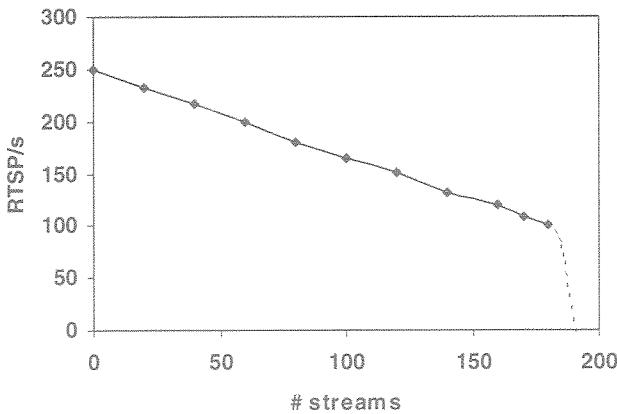


Figure 13. RTSP requests handling (AMD Athlon™ 64 processor).

(Figures 12 and 2b). The second scenario shows how the co-operative caching algorithm (Section 4.3) can efficiently create one large virtual cache, using the ‘transparent RTSP request forwarding’ principle.

5.3. Test setup and measurements

In this section, performance measurements on a prototype proxy are presented, implemented on an AMD Athlon™ 64 processor 3000+ (512 MB RAM). Figure 13 shows the number of client RTSP requests that can be handled simultaneously by the proxy, already serving RTP stream (2.5 Mbps) over a gigabit link (560 Mbps throughput measured with Iperf [26]). The proxy uses high-priority RTP threads and low-priority RTSP threads. We observe that the RTSP handling decreases linearly and fails at 190 simultaneous RTP streams (480 Mbps), due to limited system resources. Figure 14 shows the delay between PLAY request sent by a PC client and the arrival of the first RTP packet at the PC client, for different configurations (server-proxy-client). Even when the proxy has to fetch the content from the server, the delay is never higher than 35 ns (1000 measurements per configuration). When the proxy acts as a mere router, the delay caused by the server (Darwin streamer [27]) is less than 1 ms. The delay on the network links between server, proxy and client is negligible.

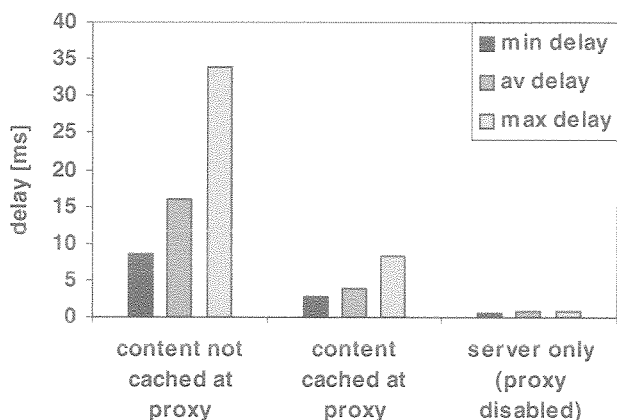


Figure 14. Delay between a client request and the actual start of the RTP stream on a client PC.

6. CONCLUSIONS

In this paper, a novel sliding-interval caching algorithm for a tsTV service was presented. Cache decisions (segment size, stored programs, ...) at low cost distributed streamers are made after each learning interval Δ , based on popularity and distance metrics. Experimental results for a basic network topology showed promising results in terms of server and network load, especially for co-operative caching. An RTSP proxy implementation has been introduced as well. A prototype, integrating the caching algorithms has been built and evaluated through measurements.

Future work includes the introduction of E2E resilience aspects (e.g. RTP retransmission [28]) and other concepts such as storage of content at the user premises, possibly served through peer-to-peer content streaming, will be investigated as well.

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