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# Indirect channels: a bandwidth-saving technique for fault-tolerant protocols

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#### Abstract

Sending large messages known to the recipient is a waste of bandwidth. Nevertheless, many fault-tolerant agreement protocols send the same large message between each pair of participating processes. This practical problem has recently been addressed in the context of Atomic Broadcast by presenting a specialized algorithm.

This paper proposes a more general solution by providing virtual *indirect channels* that physically transmit message ids instead of full messages if possible. Indirect channels are transparent to the application; they can be used with any distributed algorithm, even with unreliable channels or malicious participants. At the same time, they provide rigorous theoretical properties.

Indirect channels are conservative: they do not allow manipulating message ids if full messages are not known. This paper also investigates the consequences of relaxing this assumption on the latency and correctness of Consensus and Atomic Broadcast implementations: new algorithms and lower bounds are shown.

# 1 Introduction

Sending large network messages that are known to the receiver is a waste of bandwidth. Nevertheless, many fault-tolerant distributed protocols suffer from this problem. For example, in Consensus and Atomic Broadcast algorithms commonly perform an all-to-all message exchanges. Such an exchange involves  $O(n^2)$  messages, and results in each of the n processes receiving the same message O(n) times. The network bandwidth usage could be significantly reduced if large messages were sent only to processes that do not know them already.

This problem has been known for a long time, and many real-world protocols have taken steps to mitigate it. For example, in the Network News Transfer Protocol [11], a client can advertise the possession of a given article m with the IHAVE command. The server can then ask the client to send m or not; this avoids unnecessary communication if the server knows m already. Another example is rsync, a popular file synchronization program [1]. Before synchronizing, both end-points compute and compare hashes of parts of the file, so as to avoid transmitting parts already known to the other party. Both NNTP and rsync reduce bandwidth usage at the expense of latency: if the recipient does not know the message, three communication steps are needed instead of one ("do you have m?", "no", "here is m"). Since in point-to-point protocols, one cannot generally predict whether the recipient knows the message or not, this increase in latency is impossible to avoid.

The situation is different in group-oriented protocols, such as Atomic Broadcast, which are multi-step by nature. This means that at the beginning of the second step each process can reasonably believe that all the others have received the original broadcast as well (Figure 2). From that point on, bandwidth can be saved by sending message identifiers instead of full messages [12, 24]. Despite its apparent simplicity, this technique is surprisingly difficult to implement correctly. Great care must be taken to avoid delivering *orphaned* message ids, for which no correct process knows the corresponding full message [10]. The first such an implementation has been recently proposed by Ekwall and Schiper [10] for Atomic Broadcast.

One can imagine that a similarly careful analysis can be carried out for other distributed protocols such as Generic Broadcast, Atomic Multicast, or just other implementations of Atomic Broadcast [8]. However, the method in [10] suggests that the amount of work involved is similar to actually designing a new protocol from scratch. It would be easier if one could take any existing protocol and perform such a transformation automatically.

This paper proposes a solution. Instead of solving the problem on the protocol level, it provides a new lightweight abstraction: *indirect channels*. In brief, if the channel believes the receiver already knows the transmitted message, it sends a short message identifier instead. If this belief turns to be wrong, the receiver requests the full message from the sender. As in [10], care must be taken to avoid orphaned message identifiers.

Indirect channels differ from the NNTP/rsync synchronization method in that the decision whether to send the full message or only the id is taken locally by the sender, without consulting the recipient. This optimistic approach saves latency if the guess is right (typical), at the expense of a higher latency in anomalous runs.

Indirect channels aim at bridging the gap between theory and practice. On the one hand, they provide precise reliability guarantees required by theoretical abstractions. On the other hand, their implementation gives room to system-specific tuning, which can affects only the performance but not correctness. Cache-like memory management is a good example: all possibly large data can be discarded at any time without breaking the algorithm, with the performance degradation being the only penalty for an unlucky removal.

Standard Atomic Broadcast [6] run over indirect channels exhibits a latency similar to that of the algorithm in [10]. The advantage of indirect channels lies in the fact that they can be used with any distributed protocol without modifications, even with those tolerating malicious participants. Depending on the underlying channels, indirect channels can provide several levels of reliability guarantees. All in all, indirect channels allow an algorithm designer to separate the large-messages issue from the main problem.

By design, indirect channels conservatively avoid orphaned ids altogether. Is it possible to improve the latency even further by tolerating orphaned ids in a controlled way? Atomic Broadcast with Indirect Consensus [10] can tolerate orphaned ids in the first two steps. This approach, however, is algorithm-specific and may require more processes for the same fault resilience. I show a general method of achieving a limited orphan-tolerance with any Atomic Broadcast without decreasing its resilience. This is done by separating the agreement on id ordering from that on id-to-message mappings, and running them in parallel. Interestingly, no such improvement is possible with Consensus; this paper presents lower bounds on the maximum latency improvement resulting from treating small and large messages differently.

This paper is structured in the following way. Section 2 introduces indirect channels and presents an implementation that handles crash-failures and reliable communication. Section 2.2 extends the algorithm for unreliable communication and Byzantine failures. Section 2.3 presents an experimental evaluation. Section 3 investigates further improvements in latency of Consensus and Atomic Broadcast as a result of limited tolerance of orphaned ids.

# 2 Indirect channels

I assume that the system consists of n processes  $p_1, \ldots, p_n$ , which can fail only by crashing, and use asynchronous reliable channels for communication: all messages between correct processes *eventually* get delivered. Section 2.2 will relax these assumptions and consider eventually reliable channels and Byzantine (malicious) participants.

### 2.1 Implementation of indirect channels

The objective of this section is to implement channels that save the network bandwidth by physically transmitting a message only if the recipient does not known it. If the sender believes that the recipient knows the message, it sends the message id instead. In rare cases when this belief turns out to be wrong, the sender retransmits the full message upon a request from the recipient. To be able to fulfil that request, the sender must store its messages until the recipient acknowledges a successful reception or asks for retransmission. In the meanwhile, if the cache management demands removing this information from memory, the message must be preemptively sent to the recipient in full.

Figure 1 shows an implementation of the algorithm sketched above, which provides a reliable indirect channel using a reliable underlying channel. For clarity, I use "ind-send" and "ind-receive" for the indirect channel being implemented, and "send" and "receive" for the underlying physical channel.

I assume each message m = (s, x) consists of a collection of small fields s, and a large object x, whose transmitting is to be avoided if possible. In order to achieve this, each such object x will be assigned an identifier  $id_x$  the first time it is transmitted. Subsequent transmissions will use  $id_x$  instead of x.

Each process  $p_i$  maintains two sequence numbers:  $idm_i$  and  $idx_i$ , both initially 0. Number  $idm_i$  is the next free identifier  $id_m$  assigned to the whole message m = (s, x). Number  $idx_i$  is the next free  $id_x$  to be assigned to a large object x. Several messages m containing the same object x get different  $id_m$ 's but the same  $id_x$ . Identifiers  $id_m$  are unique locally to the issuing process,  $id_x$ 's are unique globally.

Each process also maintains the set  $mappings_i$  of all known mappings  $(id_x, x)$  from object ids  $id_x$  to actual objects x. For global uniqueness, each  $id_x$  is of the form  $(p_i, idx_i)$ , where  $p_i$  is the process that created the mapping  $(id_x, x)$ , called the *originator* of x, and  $idx_i$  is the object sequence number assigned by  $p_i$  to x.

```
idm_i \leftarrow 0; idx_i \leftarrow 0; mappings_i \leftarrow \emptyset; messages_i \leftarrow \emptyset
1
     when p_i ind-sends "m = (s, x)" to \{q_1, \ldots, q_k\} do
2
       id_m \leftarrow idm_i; increment idm_i
3
       if (id_x, x) \notin mappings_i for any id_x then
4
          id_x \leftarrow (p_i, idx_i); increment idx_i
5
          insert (id_x, x) into mappings<sub>i</sub>
6
          broadcast (MAP id_x, x)
7
       send (SHORT id_m, s, id_x) to \{q_1, \ldots, q_k\}
8
       for all p \in \{q_1, \ldots, q_k\} do
9
          insert (id_m, s, id_x, p) into messages<sub>i</sub>
10
     when p_i removes (id_m, s, id_x, p) from messages_i do
11
       let x be such that (id_x, x) \in mappings_i
12
       send (FULL id_m, s, x) to p
13
     when p_i removes (id_x, x) from mappings_i do
14
       remove all (*, *, id_x, *) from messages<sub>i</sub>
                                                                                                 \{11-13\}
15
     when p_i receives (ACK id_m) or (NACK id_m) from p do
16
       remove all (id_m, *, *, p) from messages_i
17
          if (\text{ACK } id_m) then without triggering lines 11–13
18
     when p_i receives (MAP id_x, x) do
19
       insert (id_x, x) into mappings_i
20
     when p_i receives (SHORT id_m, s, id_x) from p do
21
       wait until (id_x, x) \in mappings_i for some x
22
                     or timeout elapsed
23
            if (id_x, x) \in mappings_i then
24
               ind-receive "m = (s, x)"
25
               send (ACK id_m) to p
                                                                            {at the first opportunity}
26
            else
27
               send (NACK id_m) to p
28
     when p_i receives (FULL id_m, s, x) do
29
       ind-receive "m = (s, x)"
30
```

Figure 1: Indirect channels



Figure 2: A round of MR Consensus [18]

When a process  $p_i$  wants to ind-send a message m = (s, x) to a set of processes  $q_1, \ldots, q_k$ , it first generates a new locally unique  $id_m$  for m. Then, it checks whether  $mappings_i$  contains  $(id_x, x)$  for some  $id_x$ . If not,  $p_i$  generates a new globally unique  $id_x = (p_i, idx_i)$ , adds the new mapping  $(id_x, x)$  to  $mappings_i$ , and broadcasts it to other processes. By doing so,  $p_i$  becomes the originator of x. When a process  $p_j$  receives this mapping, it adds  $(id_x, x)$  to  $mappings_j$  (lines 19–20).

After ensuring that  $mappings_i$  contains  $(id_x, x)$ , process  $p_i$  sends  $(\text{SHORT } id_m, s, id_x)$  to processes  $q_1, \ldots, q_k$ . It also adds  $(id_m, s, id_x, p)$  to the set  $messages_i$  for all recipients  $p \in \{q_1, \ldots, q_k\}$ .

Both sets  $messages_i$  and  $mappings_i$  are caches: due to their potentially large size, individual entries can be removed at any time by the cache management system. Set  $messages_i$  consists of entries  $(id_m, s, id_x, p)$  stating that  $p_i$  has not received any acknowledgement from p for  $\langle \text{SHORT } id_m, s, id_x \rangle$  sent to it in line 8. To avoid message loss, removing such an entry requires  $p_i$  to send the full message to p (lines 11–13). When a mapping  $(id_x, x)$  is removed from  $mappings_i$ , all entries in  $messages_i$  containing  $id_x$ are removed as well, and the corresponding unacknowledged messages are sent in full (line 13).

When a process  $p_i$  receives  $\langle \text{SHORT } id_m, s, id_x \rangle$  (line 21), it waits until it knows the mapping  $(id_x, x)$ , at which point it delivers m = (s, x) and sends  $\langle \text{ACK } id_m \rangle$  to the sender. If this does not happen within the timeout period, it sends  $\langle \text{NACK } id_m \rangle$  instead. When the sender receives one of those acknowledgement (lines 16–18), it removes the corresponding entry from  $messages_i$ , if it has not been removed by the cache before. If the acknowledgement is  $\langle \text{NACK } id_m \rangle$ , this removal triggers lines 11–13, which send the corresponding full message to the recipient. Each such message is delivered directly to the application (lines 29–30).

**Behaviour in typical runs.** Figure 2 shows an execution of the Mostefaoui-Raynal (MR) Consensus algorithm [18] in a typical run. In the first step, the coordinator  $p_1$  broadcasts m = (s, x), where x is its proposal and s is some short bookkeeping information. In the second step, each process  $p_i$  broadcasts  $m_i = (s_i, x)$  where  $s_i$  is short. With normal channels, the potentially large proposal x is sent  $O(n^2)$  times.

With indirect channels,  $p_1$  first broadcast (MAP  $id_x, x$ ) and (SHORT  $id_m, s, id_x$ ) to all processes (lines 7 and 8). Each process  $p_i$  adds  $(id_x, x)$  to its mappings<sub>i</sub> (lines 19–20). In the second step, when  $p_i$  broadcasts  $m_i = (s_i, x)$ , it notices  $(id_x, x) \in mappings_i$ (line 4), so it broadcasts only (SHORT  $id_{m_i}, s_i, id_x$ ). The recipients will all eventually receive (MAP  $id_x, x$ ) from  $p_1$ , reconstruct x, and deliver  $m_i = (s_i, x)$ . Messages (ACK  $id_{m_i}$ ) will let  $p_i$  know that all processes successfully reconstructed x. Process  $p_i$  will remove the corresponding entries from  $messages_i$ , which will prevent broadcasting (FULL  $id_{m_i}, s_i, x$ ) in the future (line 13). As a result, in typical runs of Consensus, the full proposal x is broadcast only once. Note that, since each process must receive x somehow, this is the best one can achieve. This property of indirect channels, independent of the algorithm running on top, can be formalized as follows.

**Theorem 2.1 (One-reception, see A.4).** If all message delays between correct processes are shorter than  $\frac{1}{2}$ timeout, and the caches never forget information, then each correct process physically receives any large object x, originated at a correct process, at most once.

For performance reasons,  $\langle MAP \ id_x, x \rangle$  and  $\langle SHORT \ id_m, s, id_x \rangle$  in lines 7 and 8 are sent in a single message broadcast to all processes. If  $\{q_1, \ldots, q_k\} \subsetneq \{p_1, \ldots, p_n\}$ , processes  $p_j \notin \{q_1, \ldots, q_k\}$  just ignore  $\langle SHORT \rangle$ . Since  $\langle MAP \ id_x, x \rangle$  is a large message, piggybacking  $\langle SHORT \ id_m, s, id_x \rangle$  is cheap.

**Behaviour in anomalous runs.** In runs with message with delays longer than  $\frac{1}{2}$  timeout or caches forgetting their entries too quickly, indirect channels no longer guarantee onereception and can send the same object multiple times. In the worst case, when entries are removed from *messages*<sub>i</sub> and *mappings*<sub>i</sub> almost immediately, each message m = (s, x)will be transmitted in full in line 13. This is in addition to messages sent in lines 7 and 8, so both latency and bandwidth usage is higher in comparison to just using the underlying channels. Nevertheless, reliability of indirect channels is guaranteed in any run:

**Theorem 2.2 (Reliability, see A.2).** If the underlying channel is reliable, then each message ind-sent by a correct process to another correct process will eventually be ind-received.

**Duplicate messages.** If caches forget entries too quickly, indirect channels can deliver the same message twice: first in line 25 after reconstructing it from  $\langle \text{SHORT} \rangle$  and later in line 29 after receiving  $\langle \text{FULL} \rangle$  sent in line 13. In theory, such duplicates can be easily avoided by processes remembering  $id_m$ 's of received messages. Nevertheless, I believe that duplicate-suppression is usually better done at the application level, for the following reasons.

Remembering received ids requires memory that, in the worst case, grows linearly with the number of received messages. Methods of dealing with this problem are applicationspecific and include imposing a near-FIFO order by using a sliding window as in TCP [23], employing Bloom filters [5], common in anonymity systems [13], and others. Algorithms that proceed in rounds or epochs will usually discard all messages from the previous ones, so storing duplicate-information can be limited to current round. Most Consensus and Atomic Broadcast algorithms are not affected by duplicates so this problem can be ignored. Finally, Section 2.2 will show that duplicates can be eliminated altogether when one is willing to accept some message loss.

**Delaying acknowledgements.** Sending an  $\langle ACK \rangle$  immediately after receiving every message might be expensive. To mitigate this problem, a process can buffer  $\langle ACK \rangle$ s and piggyback them on other messages; most agreement protocols reply to received messages (almost) immediately (Figure 2). In general, the  $\langle ACK \rangle$  buffering delay should be small

enough to avoid spontaneous cache entry removal at the sender (lines 11–13) and resending the message in full. The more frequently messages are sent, the sooner the cache entries get removed, and the shorter the acceptable  $\langle ACK \rangle$  delay is. At the same time, high message frequency means a short waiting time for the opportunity for piggybacking an  $\langle ACK \rangle$ .

Consider a very simple model of this situation. Let d be the message delay, and f the frequency of each process broadcasting messages. With immediate  $\langle ACK \rangle$ s, each process's cache needs to be able to store each object for 2d time, which means storing 2df objects at any given time. When  $\langle ACK \rangle$ s are piggybacked on the next broadcast message (1/f wait on average), the storage requirement grows from 2df to  $(2d + 1/f) \cdot f = 2df + 1$ . This means that using delayed  $\langle ACK \rangle$ s requires the cache to be able to store only 1 more object on average. This additional amount of required space (1 object) is constant and independent on message delay or frequency.

**Cache issues.** As opposed to [10], where all mappings are kept forever, entries in  $mappings_i$  and  $messages_i$  in the algorithm in Figure 1 can be removed at any time. The strategy for entry removal is beyond the scope of this paper; the possibilities include general strategies such as LRU or LFU [22] and application-specific techniques from distributed garbage collection [2]. If the *sender* removes an entry from  $mappings_i$  too early, it might need to send more messages in lines 7 and 13. If the *receiver* removes a mapping too early, it may have to send  $\langle NACK \rangle$ s, which will also increase the latency and force the sender to broadcast full objects in lines 16–18. Removing mapping entries too late increases memory requirements for the cache.

To sum up, using a cache increases the complexity of the algorithm, however, it makes it more practical for real systems with limited memory. Most importantly, the flexibility of the cache removal strategy allows for free system-specific tuning without the fear of breaking the algorithm.

### 2.2 System extensions

So far I have assumed the crash-stop failure model and reliable channels. This section will relax those assumptions and present variants of the algorithm from Figure 1 for Byzantine settings and eventually reliable channels. Interestingly, operating with the latter model results in a significant simplification of the indirect channel implementation.

**Eventually reliable channels.** The algorithm in Figure 1 implements reliable channels, provided that the underlying channels are reliable as well. However, typical network connections are lossy, and implementing reliable channels on top of unreliable ones requires the sender to store all messages whose reception has not been confirmed. As opposed to the entries of  $mappings_i$  and  $messages_i$  in the indirect channel algorithm, this information cannot be freely discarded. As a result, a faulty recipient can always force the sender to use an unbounded amount of memory [3].

Fortunately, many distributed algorithms [8, 17] require channels that are only *eventually* reliable: the channels can be lossy until in a certain (unknown) time, after which they are permanently reliable. In practice, this period permanent reliability need not be infinite, but just long enough for the algorithm to do some useful work (in the order of several message delays) [15]. Therefore, it can be assumed that standard lossy networks satisfy this definition; in the worst case, the algorithm will stall until the network is reliable for a sufficiently long period of time.

Can the algorithm in Figure 1 be used with eventually reliable channels? Yes, Corollary A.3 shows that in if the underlying channels are eventually reliable, then the indirect channels are eventually reliable as well. Moreover, significant simplifications can be made, if all entries in mappings<sub>i</sub> are kept for at least twice a message delay plus timeout after last access. In this case, messages  $\langle ACK \rangle$  and  $\langle NACK \rangle$  will always remove elements from messages<sub>i</sub> (lines 16–18) before the cache has a chance to by executing lines 14–15. We can therefore assume that then lines 11–13 need to be executed only as a result of receiving a  $\langle NACK \rangle$  in line 15. With this observation, one can remove the set messages<sub>i</sub> entirely along with the code that references it (lines 9–15). This will require changing lines 16–18 to

- <sup>15</sup> on receive (NACK  $id_m, s, id_x$ ) from p do
- if  $(id_x, x) \in mappings_i$  for some x then
- 17 send (FULL  $id_m, s, x$ ) to p

A failure of the test in line 16 means that the information about x had been removed from  $mappings_i$ : no retransmission is possible and the message m = (s, x) will be lost.

Note that messages  $\langle NACK \rangle$  contain now more information, so line 27 needs to be changed to

27 send (NACK  $id_m, s, id_x$ )

One of the consequences of sending  $\langle \text{FULL} \rangle$  only in response to an explicit  $\langle \text{NACK} \rangle$ (line 17) is that no duplicate messages can be received. Also, since no action is taken after receiving an  $\langle \text{ACK} \rangle$ , those messages do not need to be sent at all, and line 25 can be removed. These two observations eliminate the need for explicit message duplication and  $\langle \text{ACK} \rangle$  piggybacking discussed earlier. This considerable simplification of the algorithm assumes that all entries in *mappings*<sub>i</sub> are kept long enough; if they are not, messages may get lost.

Byzantine processes and hash identifiers. Since channels are point-to-point abstractions that offer no guarantees if either endpoint is malicious, the algorithm in Figure 1 handles malicious participants with almost no modifications. The only problem is posed by processes external to the sender-receiver system, namely the originator  $p_i$  of an object, because it can convince two honest processes to map the same  $id_x$  into different objects. This problem can be solved by changing the form of  $id_x$  from  $(p_i, idx_i)$  to H(x), where H is a secure hash function. When a process receives a mapping  $(id_x, x)$  in line 19, it verifies whether  $id_x = H(x)$ , which prevents malicious processes from spreading false mappings. In fact, the new form of  $id_x$  eliminates the concept of an explicit originator from the algorithm entirely.

Hash-based message identifiers  $id_x = H(x)$  can also be useful in crash-stop settings. As opposed to  $id_x = (p_i, idx_i)$ , the same object x sent by different processes gets the same  $id_x$ . For example, if all processes propose the same x to a Consensus instance, processes can reconstruct x from received  $id_x$  without having to wait for the possible high-latency message  $\langle MAP \rangle$  broadcast in line 7. The downside of hash-based identifiers might be their size, computational cost, and the probabilistic guarantees they offer. **Dynamic groups.** The broadcast in line 7 assumes a system consisting of a fixed set of n processes. In more dynamic settings, the mapping  $\langle MAP \ id_x, x \rangle$  should be broadcast only to the intended recipients of x, direct or indirect, as determined by the application. The originator of x includes this set R of the recipients in  $id_x$ . Later, when some process re-sends x to a process  $p \notin R$ , for example if p joined the group just after the originator created  $id_x$ , then x is sent directly using  $\langle FULL \rangle$ . Such situations are atypical and should not have much impact on the overall performance.

Note on failure detectors. The algorithm in Figure 1 uses timeouts. I believe that failure detectors [6] are inadequate for implementing indirect channels because (i) they too powerful but (ii) not powerful enough.

First, classic failure detectors cannot be implemented in Byzantine settings [9]. Second, failure detectors cannot be used to implement indirect channels over eventually reliable channels. This is because per-process failure detection offered by failure detectors is not fine-grained enough to deal with individual messages being lost between correct processes (Theorem B.2). Failure detectors can therefore only be used with reliable channels.

To implement reliable indirect channels, one cannot use any of the simplifications described in earlier in this section: the original algorithm in Figure 1 must be used. This algorithm uses timeouts to ensure that processes do not wait indefinitely for messages that might never arrive (line 23). The downside of this approach is that if the timeout is too short, then unnecessary  $\langle NACK \rangle$  and  $\langle FULL \rangle$  messages are sent. This decreases the performance, but crucially, it does *not* affect the correctness of the algorithm (safety or liveness). In other words, indirect channels can be implemented correctly in purely asynchronous settings.

Now, failure detectors are too powerful to be implemented a purely asynchronous system. This is because they need to satisfy two families of properties: completeness (eventually all faulty processes suspected) and accuracy (eventually no correct processes suspected). Typical problems solved with failure detectors, such as Consensus or Atomic Broadcast, require both properties to achieve correctness. On the other hand, indirect channels require only a timeout equivalent of completeness, not accuracy. Thus, using failure detectors is overkill, and would make a wrong impression about the solvability of the problem itself.

### 2.3 Applications

Indirect channels can be used to improve throughput and latency of any distributed algorithm that tend to transmit the same object often, in particular Consensus and Atomic Broadcast protocols. For the latter case, Ekwall and Schiper [10] designed a specialized protocol and showed theoretically and experimentally that it improves both latency and throughput.

Following Ekwall and Schiper [10], I conducted an experimental evaluation of indirect channels using Neko [24], a Java framework for prototyping and evaluating distributed algorithms. Three Atomic Broadcast protocols were tested, with message payloads ranging from 100 to 10,000 bytes. Each protocol-payload combination was run for 60 seconds with three Ethernet-connected computers, one of which was constantly abcasting 50 messages with the given payload per second. The average latencies are shown in Figure 3.



Figure 3: Experimental evaluation of three Atomic Broadcast protocols

The first Atomic Broadcast algorithm is the standard Chandra and Toueg (CT) protocol [6] which embeds the payload in all its messages. The second protocol is also CT, but run over indirect channels, broadcasting each payload only once. The Neko implementation of [10] is not publicly available yet, so instead I tested a simplified but unsafe version of it, essentially CT with Consensus on message ids instead of full messages. The simplified version cannot be slower than the full algorithm [10], which is sufficient for this comparison.

**Results.** For small payload sizes, the latencies of all three protocols were similar. For the payload size over 5kB, the latency of CT started growing, whether that of the other two protocols remained relatively constant. This is because CT embeds the large payload in every message, whereas the other two algorithm broadcast the full payload only once. As a result, their latencies are dominated by small messages and are not significantly affected by the payload size.

# **3** Further improvements in latency

The indirect channel implementation in Figure 1 improves network usage by transmitting ids instead of full messages. As shown in [10] and here, this technique reduces both bandwidth usage and latency. Note that no further reduction in bandwidth is possible because each recipient must receive the full message at least once, and this is what indirect channels guarantee in typical runs (one-reception property).

The optimality of the resulting latency is not that clear. Indirect channels are conservative in that they do not allow the application to manipulate the message id before the full message is known. Without this assumption, failures in the system could force the application to decide on an *orphaned id* without a corresponding message [10]. This section will investigate whether the latency of Consensus and Atomic Broadcast can be further improved by processing short and long messages in parallel.

In the standard latency model, in which all messages have the same transmission delay d, the latency of those two abstractions is fairly well studied. Asynchronous Consensus requires at least two communication steps (2d) [7, 16] and several algorithm achieve that bound [6, 18]. Similarly, asynchronous Atomic Broadcast requires three steps (3d) [25], again with many matching algorithms [6].

I am not aware of any previous theoretical study of the impact of different sizedependent message delays on the latency of agreement protocols. This section attempts to shed some light on the problem by considering the following simple model. Assume there are only two kinds of messages: small with a transmission delay d and large with a longer delay of D > d. I will investigate the latency of Consensus and Atomic Broadcast in this model, by presenting relevant algorithms and lower bounds.

### 3.1 Consensus

In the Consensus problem, all processes propose and then agree on one of the proposals [14]. The following conditions should be met: (i) only proposals can become decisions, (ii) no two processes decide differently, (iii) all correct processes eventually decide.

Even in failure-free runs, Consensus requires two communication steps [7, 16] (Figure 2). If proposals are "large", then so are all used messages, and the resulting latency is 2D. If run over indirect channels, the second step of a Consensus algorithm [6, 18] uses short ids instead of full messages, thereby reducing the overall latency to d + D. This is an upper bound.

As for a lower bound, two communication steps are needed in any case, so the latency cannot be lower than 2d. At the same time, all processes must receive the winning proposal, which gives us a lower bound of max  $\{2d, D\}$ .

The discrepancy between those bounds  $(\max \{2d, D\} < d + D)$  poses a question whether there is a Consensus algorithm that cleverly manipulates the message ids while waiting for the full messages, and then immediately decides. The answer is "no". Theorem B.3 states that in any agreement problem, the decision value must be known, but not committed, one communication step (d units of time) in advance. Since large messages are known only after D units of time, the total latency cannot be reduced below D + d.

Indirect Consensus [10] and Consensus over indirect channels achieve the same latency, however, the latter approach works with any Consensus algorithm, without modifications. In particular, it tolerates the same number of faulty participants as in the original version. In comparison, the indirect version of MR Consensus [10] requires less than a third processes faulty (n > 3f), despite the original tolerating less than a half (n > 2f) [18]. The reason for this discrepancy is that, with indirect channels, ids without corresponding full messages do not count towards the n - f quorum in the second phase of MR, preventing orphaned ids.

### **3.2** Atomic Broadcast

In Atomic Broadcast, processes broadcast messages, and then deliver all of them in the same order [8]. Formally, we require the following properties: (i) if a correct process broadcast a message m, then all correct processes will eventually deliver m; (ii) if a process delivers a message m, then all correct processes eventually deliver m; (iii) for any message m, every process delivers m at most once, and only if m was previously abcast; and (iv) if some process delivers message m' after message m, then every process delivers m at most once, and only if m was previously abcast; and (iv) if some process delivers message m' after message m, then every process delivers m' only after it has delivered m.

Standard Atomic Broadcast protocols require three communication steps, with large messages used in all three steps (latency 3D). Running such protocols over indirect

channels limits the large messages to the first step only, thereby reducing the latency to D + 2d. This is an upper bound.

As for a lower bound, we can use the same reasoning as in Section 3.1. Three communication steps are needed (3d), and the messages must be known one step before delivery (D+d) (Theorem B.3). The resulting lower bound max  $\{3d, D+d\} < D+2d$  leaves us with the same question as before: is there an algorithm that attains it?

This time the answer is "yes" and [10] provide an example. In the second step of their Atomic Broadcast algorithm, processes can propose ids to Indirect Consensus without knowing the full messages. Achieving this with indirect channels is not straightforward because they do not allow orphaned id manipulation. One might try to modify some indirect-channel-based Consensus algorithm to allow such manipulations, but the result would be algorithm-specific.

Instead, this section shows how the latency of any three-step Atomic Broadcast algorithm can be reduced to max  $\{3d, D + d\}$ , without affecting its fault-tolerance (unlike MR-based [10]). My approach is to split the Atomic Broadcast problem into two independent problems:

- 1. Agreement on the order of delivered message ids, which can be solved by Atomic Broadcast on message ids (latency 3d).
- 2. Agreement on the mapping from message ids to full messages, which can be solved by Generic Broadcast [4, 21] over indirect channels (latency D + d).

Assuming the implementations of the two above broadcast abstractions, as well as two failure detectors:  $\Omega$  and  $\Diamond S$  [6], I will now present an Atomic Broadcast algorithm with the latency of max  $\{3d, D+d\}$  in typical runs.

#### 3.2.1 Algorithm

The details of the algorithm are shown in Figure 4. Each process  $p_i$  maintains three variables: the next message sequence number  $idm_i$ , and two sets of mappings from ids to messages: the set  $mymappings_i$  of mappings proposed by  $p_i$  and the set  $mappings_i$  of globally accepted mappings. Both sets  $mymappings_i$  and  $mappings_i$  are initially empty.

To broadcast a large message m, process  $p_i$  first generates a new unique  $id_m = (p_i, idm_i)$  and adds the mapping  $(id_m, m)$  to  $mymappings_i$ . Process  $p_i$  then broadcasts  $\langle \text{ID} \ id_m \rangle$  using Atomic Broadcast, and the mapping  $\langle \text{MAP} \ id_m, m \rangle$  using Generic Broadcast. Generic Broadcast [4, 21] is a version of Atomic Broadcast that is especially fast when messages do not *conflict*. In our case, two messages  $\langle \text{MAP} \ id_m, m \rangle$  conflict if they map the same  $id_m$  into different messages m. In typical runs, such conflicts do not occur because the  $id_m$ 's assigned by processes are unique. In runs with failures, multiple mappings with the same  $id_m$  may be sent, as we will see later.

Whenever a process  $p_i$  receives (MAP  $id_m, m$ ), it adds the mapping  $(id_m, m)$  to mappings<sub>i</sub>, unless mappings<sub>i</sub> contains a mapping for  $id_m$  already. Since all processes receive multiple mappings for a given  $id_m$  in the same order, Generic Broadcast ensures that all processes end up mapping a given  $id_m$  to the same message.

The delivery task at each process  $p_i$  consists of an infinite loop. In each iteration,  $p_i$  first waits for the  $id_m$  of a message, atomically broadcast in line 5. Then, it waits for the mapping for  $id_m$  to become known (by executing line 9). If, during this waiting,

1	$idm_i \leftarrow 0; mappings_i \leftarrow \emptyset; mymappings_i \leftarrow \emptyset$
2	when $p_i$ abcasts $m$ do
3	generate new $id_m = (p_i, idm_i)$ ; increment $idm_i$
4	insert $(id_m, m)$ into $mymappings_i$
5	broadcast $\langle \text{ID } id_m \rangle$ with Atomic Broadcast
6	broadcast $\langle \text{MAP } id_m, m \rangle$ with Generic Broadcast
7	when $p_i$ gbdelivers (MAP $id_m, m$ ) do
8	if $(id_m, m') \notin mappings_i$ for no $m'$ then
9	insert $(id_m, m)$ into $mappings_i$
10	task delivery at $p_i$ is
11	loop forever
12	wait until $(ID \ id_m)$ is abdelivered from some $p$
13	execute $\{interrupt \ lines \ 13-16 \ when \ until \ holds\}$
14 15 16	broadcast $\langle MAP \ id_m, \bot \rangle$ using Generic Broadcast until $(id_m, m) \in mappings_i$ for some $m$
17 18	$\begin{array}{l} \text{if } m \neq \bot \text{ then} \\ \text{deliver } m \end{array}$
19 20	else send $\langle \text{RETRY } id_m \rangle$ back to $p$
21	when $p_i$ receives $\langle \text{RETRY } id_m \rangle$ do
22	if $(id_m, m) \in mymappings_i$ for some $m$ then
23	remove $(id_m, m)$ from $mymappings_i$
24	$abcast(m)$ {lines 2–6}

Figure 4: Atomic Broadcast

 $p_i$  becomes the  $\Omega$ -elected leader and suspects the sender of  $id_m$ , it broadcasts an empty mapping (MAP  $id_m, \perp$ ). This ensures that  $id_m$  will be mapped to some value ( $\perp$  or m), and the potential failure of the sender of  $id_m$  will not prevent the system from delivering other messages.

When the mapping for  $id_m$  becomes finally known,  $p_i$  checks whether it is empty or not. If it is not empty, then the message m is atomically delivered. If not, the process sends a request to the sender to retry broadcasting the message (with a new  $id_m$ ). If the failure detector is  $\Diamond P$ , then each correct sender will eventually not be suspected, and its broadcasts will succeed.

With weaker failure detectors ( $\Diamond S$ ), this is not guaranteed. In this case, the sender should respond to  $\langle \text{RETRY } id_m \rangle$  by abcasting  $\langle \text{FULL } m \rangle$  using the underlying Atomic Broadcast directly, instead of executing lines 2–6. If in line 12 a process receives  $\langle \text{FULL } m \rangle$ instead of  $\langle \text{ID } id_m \rangle$ , it should deliver m straightaway and start a new iteration of the **loop**.

1	when process $p_i$ gbcasts $m$ do broadcast $m$ using Reliable Broadcast
3	when $p_i$ receives $m$ do
4 5	broadcast (CONFLICTS $m, conf_m$ ) where $conf_m$ is the sequence of previously received $m'$ conflicting with $m$
6	when $p_i$ receives $n - f$ msgs (CONFLICTS $m, conf_m$ ) do
7	if m is regular and $conf_m$ is empty then
8	deliver $m$ unless delivered before
9	else
10	for all regular $m'$ in all $conf_m$ do
11	FIFO-abcast $m'$
12	FIFO-abcast $m$
13	<b>when</b> $p_i$ abdelivers $m$ <b>do</b>
14	deliver $m$ unless delivered before

Figure 5: Two-Class Generic Broadcast

#### 3.2.2 Latency analysis

Consider the latency of the algorithm in runs in which no process fails during abcasting its messages. In such runs, Atomic Broadcast of  $\langle \text{ID} id_m \rangle$  takes 3d units of time, and Generic Broadcast of  $\langle \text{MAP} id_m, m \rangle$  over indirect channels takes D + d because the only possible conflicting mapping  $\langle \text{MAP} id_m, \perp \rangle$  is never sent. As a result, all messages m are atomically delivered in max  $\{3d, D + d\}$  time. This is a lower bound, as shown before.

One problem with the above analysis is that Generic Broadcast can be implemented with a two-step latency only if less than a third of processes are faulty (n > 3f) [20]. Ekwall and Schiper [10] report the same requirement in order to use a two-step MR Consensus algorithm [18]. Is the condition n > 3f necessary for achieving the max  $\{3d, D + d\}$ latency?

No. This section shows a variant of Generic Broadcast that delivers messages in two steps while requiring only n > 2f. Messages  $\langle MAP \rangle$  in Figure 4 can be divided into two classes: regular  $\langle MAP \ id_m, m \rangle$  and special  $\langle MAP \ id_m, \perp \rangle$ . Note that: (i) regular messages never conflict with each other, and (ii) only regular messages need to be delivered quickly. Another example of regular and special messages can be read and write requests in systems in which writes are rare.

Figure 5 shows an implementation of Two-Class Generic Broadcast. To broadcast a message m, a process sends it using reliable broadcast [15]. Upon receiving m, each process rebroadcasts it along with the sequence  $conf_m$  of previously received messages conflicting with m. When a process receives n - f such messages, it first checks whether m is regular and all  $conf_m$ 's are empty, and delivers m if so. Otherwise, all regular messages in all  $conf_m$ 's are broadcast using an underlying FIFO Atomic Broadcast protocol, followed by m itself. All duplicates are explicitly removed on delivery.

Two messages  $m_1$  and  $m_2$  can conflict only if one of them is special. They can therefore be delivered in different orders at different processes only if (i) exactly one of them, say  $m_1$ , is regular and is delivered in line 8, and (ii) some process abcasts  $m_2$  in line 12 without abcasting  $m_1$  in line 11. The first means that n - f processes received  $m_1$  before  $m_2$ , the second that n - f processes received  $m_2$  before  $m_1$ . This contradicts the assumption that n > 2f. Note that no Atomic Broadcast is used in the absence of conflicting messages.

Using Two-Class Broadcast over indirect channels, allows one to reduce the latency of any three-step Atomic Broadcast implementation to max  $\{3d, D + d\}$  without reducing its fault-tolerance. This should be contrasted with the indirect version of MR-based Atomic Broadcast, which reduces fault-tolerance from n > 2f to n > 3f [10].

### **3.3** Other broadcast abstractions

Optimistic Atomic Broadcast [19] and Generic Broadcast [4, 21] require two communication steps (2D) in runs with spontaneous order and no conflicts, respectively. Using indirect channels, the latency can be reduced to D + d, which is also a lower bound by Theorem B.3.

# 4 Conclusion

Excessive bandwidth usage caused sending the same large messages several times can be reduced by transmitting their ids whenever possible. Although this method has been widely used in many real-world point-to-point protocols [1, 11], applying it to fault-tolerant group protocols is surprisingly difficult [10]. For both families, the existing solutions are problem-specific.

This paper proposed a more general solution by providing virtual *indirect channels* that physically send a full message only during its first transmission; all other transmission use the message id only. Indirect channels are transparent to the application and have no latency overhead in typical runs. As a result, they can be used with any distributed algorithm, without modifications, even with unreliable channels and malicious participants. While providing rigorous theoretical properties at all times, the implementation lends itself to system-specific tuning through cache configuration, making it attractive for practical systems.

Running Consensus algorithms over indirect channels produces protocols with optimum latency and fault-tolerance. In the case of Atomic Broadcast, the latency of any implementing protocol can be further improved by handling message ordering and id mapping separately. The resulting latency and fault-tolerance are lower bounds, however, the problem whether this can be achieved just by using more sophisticated indirect channels is an open question.

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# A Correctness proofs

## A.1 Indirect channels

**Lemma A.1.** If  $(id_m, s, id_x, p) \in messages_i$  then  $(id_x, m) \in mappings_i$  for some m.

*Proof.* True at the beginning, we need to prove that every action preserves this property. Removing  $(id_x, x)$  from  $mappings_i$  removes all  $(*, *, id_x, *)$  from  $messages_i$  (line 15). Adding  $(id_m, s, id_x, p)$  to  $messages_i$  can only happen after ensuring that  $(id_x, x) \in mappings_i$  (line 10).

**Theorem A.2 (Reliability).** If the underlying channel is reliable, then each message m = (s, x) ind-sent by a correct process  $p_i$  to another correct process  $p_j$  will eventually be ind-received.

Proof. Executing ind-send will add  $(id_m, s, id_x, p_j)$  to  $messages_i$ , and send  $\langle \text{SHORT } id_m, s, id_x \rangle$  to  $p_j$ , which  $p_j$  will eventually receive. Because of the timeout involved, wait in lines 22–23 will eventually terminate. If  $(id_x, x) \in mappings_j$ , then m = (s, x) will be delivered. Otherwise,  $\langle \text{NACK } id_m \rangle$  will be sent back to  $p_i$ , which will remove  $(id_m, s, id_x, p_j)$  from  $messages_i$  if it is there. Therefore,  $(id_m, s, id_x, p_j)$  is at some point removed from  $messages_i$ , which involves sending  $\langle \text{FULL } id_m, s, x \rangle$  to  $p_j$  (line 13), so  $p_j$  will ind-receive it. If lines 11–13 were not triggered, then  $p_i$  received  $\langle \text{ACK } id_m \rangle$  from  $p_j$  (line 18), which means that  $p_j$  ind-received m = (s, x).

**Corollary A.3 (Eventual reliability).** If the underlying channel is eventually reliable, eventually each message m indirect-sent by a correct process to another correct process will eventually be ind-received.

**Theorem A.4 (One-reception).** If all message delays d between correct processes satisfy 2d < timeout, and the caches never forget information, then each correct process physically receives any large object x, originated at a correct process, at most once.

*Proof.* It is sufficient to to show that the only time object x is physically sent is by the originator  $p_k$  in line 7, as part of  $\langle MAP \ id_x, x \rangle$ . This mapping will be in  $mappings_k$  forever. No other process  $p_i$  can send x without indirect-receiving it first in line 26, which requires  $(id_x, x) \in mappings_i$ . Therefore, any subsequent execution of ind-send will encounter  $(id_x, x) \in mappings_i$  and line 7 will not be executed.

Assume the originator broadcasts the initial  $\langle MAP \ id_x, x \rangle$  at time 0. All correct processes receive this mapping by time d. Any  $id_x$  sent in line 8 at time  $t \ge 0$ , will arrive at the recipient p at a time t' between t and t+d. Since t' + timeout > t+2d > d, lines 22–23 will never time out. Similarly, max  $\{t', d\} \le t+d$ , so the  $\langle ACK \ id_x \rangle$  will arrive back at the sender by time t + 2d < t + timeout, and  $(id_m, s, id_x, p)$  will be removed from mappings without triggering lines 11–13. No  $\langle NACK \rangle$ s will ever be sent and caches hold information long enough, so no removal attempt will be made before. As a result, line 13 will never be executed.

**Theorem A.5 (Byzantine validity).** If an honest process  $p_i$  ind-received m = (s, x), and the sender is honest, then it has indirect-sent m = (s, x).

Proof. The sender must have sent  $\langle \text{SHORT } id_m, s, id_x \rangle$  such that  $(id_x, x) \in mappings_i$ . An honest process puts  $(id_x, x)$  in mappings either in line 6 or in line 20, both cases ensuring that  $id_x = H(x)$ . Assuming no hash collisions, two honest processes cannot resolve the same  $id_x$  to a different message. If an honest process ind-receives m = (s, x)from another honest process, then the sender must have sent  $\langle \text{SHORT } id_m, s, id_x \rangle$  in line 8, after executing ind-send m in line 2.

### A.2 Atomic Broadcast

**Lemma A.6.** If  $(id_m, m_i) \in mappings_i$  and  $(id_m, m_j) \in mappings_j$ , then  $m_i = m_j$ .

Proof. If  $m_i \neq m_j$ , then the assumption implies that  $p_i$  gbdelivered  $(id_m, m_i)$  before  $(id_m, m_j)$ , but  $p_j$  gbdelivered  $(id_m, m_j)$  before  $(id_m, m_i)$ . Since  $(id_m, m_j)$  conflicts with  $(id_m, m_i)$ , this contradicts the Uniform Partial Order property of the underlying Generic Broadcast.

Lemma A.7. Lines 13–16 will eventually terminate at any correct process.

*Proof.* The Total Order property of the underlying Atomic Broadcast implies that all processes abdeliver  $id_m$ 's in line 12 in the same order. For the sake of contradiction, consider the first  $id_m$  for which lines 13–16 do not terminate at some correct process  $p_i$ .

If the sender  $p_j$  is correct, then it has gbcast  $\langle MAP \ id_m, m \rangle$  in line 6, so eventually  $(id_m, m') \in mappings_i$  for some m', which implies the assertion. Therefore, assume  $p_j$  is faulty. By the choice of  $id_m$  and Uniform Agreement of Atomic Broadcast, all correct processes executed line 12 for  $id_m$ . By the choice of  $id_m$  and Uniform Agreement of Generic Broadcast, lines 13–16 will terminate for  $id_m$ ,  $\perp \rangle$  in line 15. By Validity of Generic Broadcast, lines 13–16 will terminate at all correct processes, which proves the assertion.

**Theorem A.8 (Validity).** If a correct process broadcast a message m, then all correct processes will eventually deliver m.

*Proof.* Validity of the underlying Atomic Broadcast implies that all correct processes will eventually abdeliver  $id_m$  in line 12. Lemma A.7 ensures that lines 13–16 will eventually terminate. If no process suspects the sender, then  $\langle MAP \ id_m, \perp \rangle$  has never been gbcast in line 15, and Integrity of Generic Broadcast implies that  $\langle MAP \ id_m, m \rangle$  gbcast in line 6 is the only possible in line 7. Therefore  $(id_m, m) \in mapping_i$  in line 17, and m is delivered.

If the sender is suspected, then the condition in line 17 might not hold. In this case, the sender will receive  $\langle \text{RETRY } id_m \rangle$  in lines 21–24 and repeat the abcasting procedure in lines 2–6, with a new  $id_m$ . Eventual Accuracy of  $\Diamond P$  ensures that a correct sender will eventually not be suspected, so some repetition of lines 2–6 will eventually succeed in atomically delivering m in line 18.

**Theorem A.9 (Uniform Agreement).** If a process  $p_i$  delivers a message m, then all correct processes eventually deliver m.

Proof. If  $p_i$  abdelivers  $id_{m_1}, id_{m_2}, \ldots, id_m$  in line 12, then Lemma A.7 and Uniform Agreement of the underlying Atomic Broadcast ensure that all correct processes will do so as well. Since  $(id_m, m) \in mappings_i$ , Uniform Agreement of Generic Broadcast and Lemma A.6 ensure that eventually  $(id_m, m) \in mappings_j$  at all correct processes  $p_j$ . This implies the assertion.

**Theorem A.10 (Uniform Integrity).** For any message m, every process delivers m at most once, and only if m was previously abcast.

Proof. Let p be the abcaster of m and  $id_1, \ldots$  be ids assigned to m by p in line 4. Any  $id_{k+1}$  is assigned to m as a result of p receiving "retry  $id_m$ ", which is sent only if  $(id_k, \bot) \in mappings_i$  at some process  $p_i$ . By Lemma A.6  $(id_k, m)$  cannot hold at any process  $p_i$ . The only  $id_k$  for which  $(id_k, m) \in mappings_i$  can hold is the last in the sequence  $id_1, \ldots$  (if it exist). This proves the first part of the assertion.

For the second part, if m is delivered (line 18), then  $(id_m, m) \in mappings_i$  for some  $id_m$ . This means that  $(MAP \ id_m, m)$  was gbcast in line 6, which implies the conclusion.

**Theorem A.11 (Uniform Total Order).** If some process  $p_i$  delivers message m' after message m, then every process  $p_i$  delivers m' only after it has delivered m.

Proof. Let  $id_{m_1}, ..., id_m, ..., id_{m'}, ...$  be the sequence of ids delivered by process  $p_i$  in line 12. Process  $p_j$  delivered  $id_{m'}$ , so Uniform Total Order of the underlying Atomic Broadcast implies that  $p_j$  must have delivered  $id_m$ , which then passed the test  $(id_m, m'') \in mappings_j$ in line 16 for some m''. Since  $(id_m, m) \in mappings_i$ , Lemma A.6 implies m'' = m, which implies the conclusion.

### A.3 Two-class Broadcast

**Lemma A.12.** If all n - f correct processes receive m in line 3, then all correct processes will eventually deliver m.

*Proof.* Those processes broadcast  $(\text{CONFLICT } m, conf_m)$ , so all correct processes will execute lines 6–12. If m is regular and all  $conf_m$  at all correct processes are empty, then all of them will deliver m in line 8. Otherwise, some correct process will abcast m in line 12, which will eventually be delivered by all correct processes.

**Theorem A.13 (Validity).** If a correct process broadcasts a message m, then all correct processes will eventually deliver m.

*Proof.* The assumption implies that all n - f correct processes will receive m; the conclusion follows from Lemma A.12.

**Theorem A.14 (Uniform Agreement).** If a process delivers a message m, then all correct processes eventually deliver m.

*Proof.* If the delivery occurs in line 14, then the assertion follows from Uniform Agreement of the underlying Atomic Broadcast. If the delivery occurs in line 8, then n - f processes broadcast (CONFLICT  $m, conf_m$ ), at least one of them correct (n > 2f). Reliable broadcast used in line 2 ensures that eventually all n - f correct processes will receive m. Lemma A.12 implies the assertion.

**Theorem A.15 (Uniform Integrity).** For any message m, every process delivers m at most once, and only if m was previously abcast.

*Proof.* Explicit duplicate elimination ensures the first part of the assertion. For the second part, if m is delivered, some process must have received  $(\text{CONFLICTS } m, conf_m)$  in line 6, so some process must have received m in line 3, which implies the conclusion.

**Theorem A.16 (Uniform Partial Order).** If some process delivers message  $m_2$  after a conflicting message  $m_1$ , then every process delivers  $m_2$  only after it has delivered  $m_1$ .

Proof. If none of those messages is delivered in line 8, then the conclusion follows from Uniform Total Order of the underlying Atomic Broadcast. If one of the messages (m) is delivered in line 8, then it is regular, and the other one (m') must be special (because regular messages do not conflict). Moreover, at least n - f processes must have received m before m' (empty  $conf_m$ ), so each process abcasting m' has at least one  $m \in conf_{m'}$ , so it abcasts m before. The FIFO property implies that all processes deliver m before m'.

# **B** Impossibility results

**Definition B.1 (One-reception for failure detectors).** If all processes are correct, there are no suspicions, and the caches never forget information, then each correct process physically receives any large object x at most once.

**Theorem B.2.** If the underlying channel can lose messages, then one-reception cannot be achieved with standard failure detectors.

*Proof.* Assume all caches hold all information forever, all processes are correct, and none are ever suspected. Consider a run in one process p ind-broadcasts a large message m every second or so. In this run, the actual message m, which should be physically broadcast at most once, arrives at all other processes at time t. Call this run r(t).

Now consider another run r' in which this single broadcast is lost; note that r' does not depend on t. To guarantee eventual reliability, all processes will eventually, say at time t', have to start delivering messages m broadcast by p. Therefore, all processes will have to receive m by time t'. Consider any run r(t) with t > t'. These runs are indistinguishable until time t, so they both receive m at time t', but r(t) also receives m at time t > t'. This violates the one-reception property.

**Theorem B.3.** Time D + d is required for asynchronous agreement protocols.

*Proof.* To obtain contradiction, consider an algorithm that decides before D + d in every execution in which all processes are correct and there is no message loss. With broadcast protocols, assume that process  $p_1$  broadcasts  $m_1$  at time 0, and no other broadcast takes place. For Consensus, assume each process  $p_i$  proposes a different value  $m_i$  at time 0; without loss of generality we can assume that all processes decide on  $m_1$ . For both, broadcasts and Consensus, the decision (delivery) must happen before time D + d.

Consider a scenario in which  $p_1$  is faulty, and all large messages sent by  $p_1$  to other processes are lost (but no others). This scenario is indistinguishable from the previous one

to processes other than  $p_1$  until time D, and to process  $p_1$  until time D + d. As a result, process  $p_1$  decides on  $m_1$  before time D + d, and say crashes at time D + d. No other process will ever receive  $m_1$  but it will be required to decide on it by Uniform Agreement. This is a contradiction.