Improved Packet Reordering Metrics

Status of This Memo

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IESG Note

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Abstract

This document presents two improved metrics for packet reordering, namely, Reorder Density (RD) and Reorder Buffer-occupancy Density (RBD). A threshold is used to clearly define when a packet is considered lost, to bound computational complexity at O(N), and to keep the memory requirement for evaluation independent of N, where N is the length of the packet sequence. RD is a comprehensive metric that captures the characteristics of reordering, while RBD evaluates the sequences from the point of view of recovery from reordering.
These metrics are simple to compute yet comprehensive in their characterization of packet reordering. The measures are robust and orthogonal to packet loss and duplication.

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1. Introduction and Motivation

Packet reordering is a phenomenon that occurs in Internet Protocol (IP) networks. Major causes of packet reordering include, but are not limited to, packet striping at layers 2 and 3 [Ben99] [Jai03], priority scheduling (e.g., Diffserv), and route fluttering [Pax97] [Boh03]. Reordering leads to degradation of the performance of many applications [Ben99] [Bla02] [Lao02]. Increased link speeds [Bar04], increased parallelism within routers and switches, Quality-of-Service (QoS) support, and load balancing among links all point to increased packet reordering in future networks.

Effective metrics for reordering are required to measure and quantify reordering. A good metric or a set of metrics capturing the nature of reordering can be expected to provide insight into the reordering phenomenon in networks. It may be possible to use such metrics to predict the effects of reordering on applications that are sensitive to packet reordering, and perhaps even to compensate for reordering. A metric for reordered packets may also help evaluate network protocols and implementations with respect to their impact on packet reordering.

The percentage of out-of-order packets is often used as a metric for characterizing reordering. However, this metric is vague and lacking in detail. Further, there is no uniform definition for the degree of reordering of an arrived packet [Ban02] [Pi05a]. For example, consider the two packet sequences (1, 3, 4, 2, 5) and (1, 4, 3, 2, 5). It is possible to interpret the reordering of packets in these sequences differently. For example,

(i) Packets 2, 3, and 4 are out of order in both cases.

(ii) Only packet 2 is out of order in the first sequence, while packets 2 and 3 are out of order in the second.

(iii) Packets 3 and 4 are out of order in both the sequences.

(iv) Packets 2, 3, and 4 are out of order in the first sequence, while packets 4 and 2 are out of order in the second sequence.

In essence, the percentage of out-of-order packets as a metric of reordering is subject to interpretation and cannot capture the reordering unambiguously and hence, accurately.

Other metrics attempt to overcome this ambiguity by defining only the late packets or only the early packets as being reordered. However, measuring reordering based only on late or only on early packets is not always effective. Consider, for example, the sequence (1, 20, 2,
3, ..., 19, 21, 22, ...); the only anomaly is that packet 20 is
delivered immediately after packet 1. A metric based only on
lateness will indicate a high degree of reordering, even though in
this example it is a single packet arriving ahead of others.
Similarly, a metric based only on earliness does not accurately
capture reordering caused by a late arriving packet. A complete
reorder metric must account for both earliness and lateness, and it
must be able to differentiate between the two. The inability to
capture both the earliness and the lateness precludes a metric from
being useful for estimating end-to-end reordering based on reordering
in constituent subnets.

The sensitivity to packet reordering can vary significantly from one
application to the other. Consider again the packet sequence (1, 3,
4, 2, 5). If buffers are available to store packets 3 and 4 while
waiting for packet 2, an application can recover from reordering.
However, with certain real-time applications, the out-of-order
arrival of packet 2 may render it useless. While one can argue that
a good packet reordering measurement scheme should capture
application-specific effects, a counter argument can also be made
that packet reordering should be measured strictly with respect to
the order of delivery, independent of the application.

Many different packet reordering metrics have been suggested. For
example, the standards-track document RFC 4737 [RFC4737] defines 11
metrics for packet reordering, including lateness-based percentage
metrics, reordering extent metrics, and N-reordering.

Section 2 of this document discusses the desirable attributes of any
packet reordering metric. Section 3 introduces two additional packet
reorder metrics: Reorder Density (RD) and Reorder Buffer-occupancy
Density (RBD), which we claim are superior to the others [Pi07]. In
particular, RD possesses all the desirable attributes, while other
metrics fall significantly short in several of these attributes. RBD
is unique in measuring reordering in terms of the system resources
needed for recovery from packet reordering. Both RD and RBD have a
computation complexity O(N), where N is the length of the packet
sequence, and they can therefore be used for real-time online
monitoring.

2. Attributes of Packet Reordering Metrics

The first and foremost requirement of a packet reordering metric is
its ability to capture the amount and extent of reordering in a
sequence of packets. The fact that a measure varies with reordering
of packets in a stream does not make it a good metric. In [Ben99],
the authors have identified desirable features of a reordering
metric. This list encloses the foremost requirements stated above:
simplicity, low sensitivity to packet loss, ability to combine reorder measures from two networks, minimal value for in-order data, and independence of data size. These features are explained below in detail, along with additional desired features. Note, the ability to combine reorder measures from two networks is added to broaden applicability, and data size independence is discussed under evaluation complexity. However, data size independence could also refer to the final measure, as in percentage reordering or even a normalized representation.

a) Simplicity

An ideal metric is one that is simple to understand and evaluate, and yet informative, i.e., able to provide a complete picture of reordering. Percentage of packets reordered is the simplest singleton metric; but the ambiguity in its definition, as discussed earlier, and its failure to carry the extent of reordering make it less informative. On the other hand, keeping track of the displacements of each and every packet without compressing the data will contain all the information about reordering, but it is not simple to evaluate or use.

A simpler metric may be preferred in some cases even though it does not capture reordering completely, while other cases may demand a more complex, yet complete metric.

In striving to strike a balance, the lateness-based metrics consider only the late packets as reordered, and earliness-based metrics only the early packets as reordered. However, a metric based only on earliness or only on lateness captures only a part of the information associated with reordering. In contrast, a metric capturing both early and late arrivals provides a complete picture of reordering in a sequence.

b) Low Sensitivity to Packet Loss and Duplication

A reorder metric should treat only an out-of-order packet as reordered, i.e., if a packet is lost during transit, then this should not result in its following packets, which arrive in order, being classified as out of order. Consider the sequence (1, 3, 4, 5, 6). If packet 2 has been lost, the sequence should not be considered to contain any out-of-order packets. Similarly, if multiple copies of a packet (duplicates) are delivered, this must
not result in a packet being classified as out of order, as long as one copy arrives in the proper position. For example, sequence (1, 2, 3, 2, 4, 5) has no reordering. The lost and duplicate packet counts may be tracked using metrics specifically intended to measure those, e.g., percentage of lost packets, and percentage of duplicate packets.

c) Low Evaluation Complexity

Memory and time complexities associated with evaluating a metric play a vital role in implementation and real-time measurements. Spatial/memory complexity corresponds to the amount of buffers required for the overall measurement process, whereas time/computation complexity refers to the number of computation steps involved in computing the amount of reordering in a sequence. On-the-fly evaluation of the metric for large streams of packets requires the computational complexity to be $O(N)$, where $N$ denotes the number of received packets, used for the reordering measure. This allows the metric to be updated in constant-time as each packet arrives. In the absence of a threshold defining losses or the number of sequence numbers to buffer for detection of duplicates, the worst-case complexity of loss and duplication detection will increase with $N$. The rate of increase will depend, among other things, on the value of $N$ and the implementation of the duplicate detection scheme.

d) Robustness

Reorder measurements should be robust against different network phenomena and peculiarities in measurement or sequences such as a very late arrival of a duplicate packet, or even a rogue packet due to an error or sequence number wraparound. The impact due to an event associated with a single or a small number of packets should have a sense of proportionality on the reorder measure. Consider, for example, the arrival sequence: (1, 5430, 2, 3, 4, 5, ...) where packet 5430 appears to be very early; it may be due to either sequence rollover in test streams or some unknown reason.

e) Broad Applicability

A framework for IP performance metrics [RFC2330] states: "The metrics must aid users and providers in understanding the performance they experience or provide".

Rather than being a mere value or a set of values that changes with the reordering of packets in a stream, a reorder metric should be useful for a variety of purposes. An application or a transport protocol implementation, for example, may be able to use
the reordering information to allocate resources to recover from reordering. A metric may be useful for TCP flow control, buffer resource allocation for recovery from reordering and/or network diagnosis.

The ability to combine the reorder metrics of constituent subnets to measure the end-to-end reordering would be an extremely useful property. In the absence of this property, no amount of individual network measurements, short of measuring the reordering for the pair of endpoints of interest, would be useful in predicting the end-to-end reordering.

The ability to provide different types of information based on monitoring or diagnostic needs also broadens the applicability of a metric. Examples of applicable information for reordering may include parameters such as the percentage of reordered packets that resulted in fast retransmissions in TCP, or the percentage of utilization of the reorder recovery buffer.

3. Reorder Density and Reorder Buffer-Occupancy Density

In this memo, we define two discrete density functions, Reorder Density (RD) and Reorder Buffer-occupancy Density (RBD), that capture the nature of reordering in a packet stream. These two metrics can be used individually or collectively to characterize the reordering in a packet stream. Also presented are algorithms for real-time evaluation of these metrics for an incoming packet stream.

RD is defined as the distribution of displacements of packets from their original positions, normalized with respect to the number of packets. An early packet corresponds to a negative displacement and a late packet to a positive displacement. A threshold on displacement is used to keep the computation within bounds. The choice of threshold value depends on the measurement uses and constraints, such as whether duplicate packets are accounted for when evaluating these displacements (discussed in Section 5).

The ability of RD to capture the nature and properties of reordering in a comprehensive manner has been demonstrated in [Pi05a], [Pi05b], [Pi05c], and [Pi07]. The RD observed at the output port of a subnet when the input is an in-order packet stream can be viewed as a "reorder response" of a network, a concept somewhat similar to the "system response" or "impulse response" used in traditional system theory. For a subnet under stationary conditions, RD is the probability density of the packet displacement. RD measured on individual subnets can be combined, using the convolution operation, to predict the end-to-end reorder characteristics of the network formed by the cascade of subnets under a fairly broad set of
conditions [Pi05b]. RD also shows significant promise as a tool for analytical modeling of reordering, as demonstrated with a load-balancing scenario in [Pi06]. Use of a threshold to define the condition under which a packet is considered lost makes the metric robust, efficient, and adaptable for different network and stream characteristics.

RBD is the normalized histogram of the occupancy of a hypothetical buffer that would allow the recovery from out-of-order delivery of packets. If an arriving packet is early, it is added to a hypothetical buffer until it can be released in order [Ban02]. The occupancy of this buffer, after each arrival, is used as the measure of reordering. A threshold, used to declare a packet as lost, keeps the complexity of computation within bounds. The threshold may be selected based on application requirements in situations where the late arrival of a packet makes it useless, e.g., a real-time application. In [Ban02], this metric was called RD and buffer occupancy was known as displacement.

RD and RBD are simple, yet useful, metrics for measurement and evaluation of reordering. These metrics are robust against many peculiarities, such as those discussed previously, and have a computational complexity of O(N), where N is the received sequence size. RD is orthogonal to loss and duplication, whereas RBD is orthogonal to duplication.

A detailed comparison of these and other proposed metrics for reordering is presented in [Pi07].

The following terms are used to formally define RD, RBD, and the measurement algorithms. The wraparound of sequence numbers is not addressed in this document explicitly, but with the use of modulo-N arithmetic, all claims made here remain valid in the presence of wraparound.

3.1. Receive Index (RI)

Consider a sequence of packets (1, 2, ..., N) transmitted over a network. A receive index RI (1, 2, ...) is a value assigned to a packet as it arrives at its destination, according to the order of arrival. A receive index is not assigned to duplicate packets, and the receive index value skips the value corresponding to a lost packet. (The detection of loss and duplication for this purpose is described in Section 6.) In the absence of reordering, the sequence number of the packet and the receive index are the same for each packet.
RI is used to compute earliness and lateness of an arriving packet. Below are two examples of received sequences with receive index values for a sequence of 5 packets (1, 2, 3, 4, 5) arriving out of order:

Example 1:
Arrived sequence:    2   1   4   5    3
receive index:       1   2   3   4    5

Example 2:
Arrived sequence:    1   4   3   5    3
receive index:       1   3   4   5    -

In Example 1, there is no loss or duplication. In Example 2, the packet with sequence number 2 is lost. Thus, 2 is not assigned as an RI. Packet 3 is duplicated; thus, the second copy is not assigned an RI.

3.2. Out-of-Order Packet

When the sequence number of a packet is not equal to the RI assigned to it, it is considered to be an out-of-order packet. Duplicates for which an RI is not defined are ignored.

3.3. Displacement (D)

Displacement (D) of a packet is defined as the difference between RI and the sequence number of the packet, i.e., the displacement of packet i is RI[i] - i. Thus, a negative displacement indicates the earliness of a packet and a positive displacement the lateness. In example 3 below, an arrived sequence with displacements of each packet is illustrated.

Example 3:
Arrived sequence:    1   4   3   5   3   8   7   6
receive index:       1   3   4   5   -   6   7   8
Displacement:        0  -1   1   0   -  -2   0   2

3.4. Displacement Threshold (DT)

The displacement threshold is a threshold on the displacement of packets that allows the metric to classify a packet as lost or duplicate. Determining when to classify a packet as lost is difficult because there is no point in time at which a packet can definitely be classified as lost; the packet may still arrive after some arbitrarily long delay. However, from a practical point of view, a packet may be classified as lost if it has not arrived within a certain administratively defined displacement threshold, DT.
Similarly, to identify a duplicate packet, it is theoretically necessary to keep track of all the arrived (or missing) packets. Again, however, from a practical point of view, missing packets within a certain window of sequence numbers suffice. Thus, DT is used as a practical means for declaring a packet as lost or duplicated. DT makes the metric more robust, keeps the computational complexity for long sequences within $O(N)$, and keeps storage requirements independent of $N$.

If the DT selected is too small, reordered packets might be classified as lost. A large DT will increase both the size of memory required to keep track of sequence numbers and the length of computation time required to evaluate the metric. Indeed, it is possible to use two different thresholds for the two cases. The selection of DT is further discussed in Section 5.

3.5. Displacement Frequency (FD)

Displacement Frequency $FD[k]$ is the number of arrived packets having a displacement of $k$, where $k$ takes values from $-DT$ to $DT$.

3.6. Reorder Density (RD)

RD is defined as the distribution of the Displacement Frequencies $FD[k]$, normalized with respect to $N'$, where $N'$ is the length of the received sequence, ignoring lost and duplicate packets. $N'$ is equal to the sum($FD[k]$) for $k$ in $[-DT, DT]$.

3.7. Expected Packet (E)

A packet with sequence number $E$ is expected if $E$ is the largest number such that all the packets with sequence numbers less than $E$ have already arrived or have been determined to be lost.

3.8. Buffer Occupancy (B)

An arrived packet with a sequence number greater than that of an expected packet is considered to be stored in a hypothetical buffer sufficiently long to permit recovery from reordering. At any packet arrival instant, the buffer occupancy is equal to the number of out-of-order packets in the buffer, including the newly arrived packet. One buffer location is assumed for each packet, although it is possible to extend the concept to the case where the number of bytes is used for buffer occupancy. For example, consider the
sequence of packets (1, 2, 4, 5, 3) with expected order (1, 2, 3, 4, 5). When packet 4 arrives, the buffer occupancy is 1 because packet 4 arrived early. Similarly, the buffer occupancy becomes 2 when packet 5 arrives. When packet 3 arrives, recovery from reordering occurs and the buffer occupancy reduces to zero.

3.9. Buffer-Occupancy Threshold (BT)

Buffer-occupancy threshold is a threshold on the maximum size of the hypothetical buffer that is used for recovery from reordering. As with the case of DT for RD, BT is used for loss and duplication classification for Reorder Buffer-occupancy Density (RBD) computation (see Section 3.11). BT provides robustness and limits the computational complexity of RBD.

3.10. Buffer-Occupancy Frequency (FB)

At the arrival of each packet, the buffer occupancy may take any value, k, ranging from 0 to BT. The buffer occupancy frequency FB[k] is the number of arrival instances after which the occupancy takes the value of k.

3.11. Reorder Buffer-Occupancy Density (RBD)

Reorder buffer-occupancy density is the buffer occupancy frequencies normalized by the total number of non-duplicate packets, i.e., RBD[k] = FB[k]/N' where N' is the length of the received sequence, ignoring excessively delayed (deemed lost) and duplicate packets. N' is also the sum(FB[k]) for all k such that k belongs to [0, BT].

4. Representation of Packet Reordering and Reorder Density

Consider a sequence of packets (1, 2, ..., N). Let the RI assigned to packet m be "the sequence number m plus an offset dm", i.e.,

RI = m + dm; D = dm

A reorder event of packet m is represented by r(m, dm). When dm is not equal to zero, a reorder event is said to have occurred. A packet is late if dm > 0 and early if dm < 0. Thus, packet reordering of a sequence of packets is completely represented by the union of reorder events, R, referred to as the reorder set:

R = {r(m, dm) | dm not equal to 0 for all m}

If there is no reordering in a packet sequence, then R is the null set.
Examples 4 and 5 illustrate the reorder set:

Example 4. No losses or duplicates

<table>
<thead>
<tr>
<th>Arrived Sequence</th>
<th>1</th>
<th>2</th>
<th>3</th>
<th>5</th>
<th>4</th>
<th>6</th>
</tr>
</thead>
<tbody>
<tr>
<td>receive index (RI)</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>5</td>
<td>6</td>
</tr>
<tr>
<td>Displacement (D)</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>-1</td>
<td>1</td>
<td>0</td>
</tr>
<tr>
<td>R</td>
<td>{(4,1), (5,-1)}</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Example 5. Packet 4 is lost and 2 is duplicated

<table>
<thead>
<tr>
<th>Arrived Sequence</th>
<th>1</th>
<th>2</th>
<th>5</th>
<th>3</th>
<th>6</th>
<th>2</th>
</tr>
</thead>
<tbody>
<tr>
<td>receive index (RI)</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>5</td>
<td>6</td>
<td>-</td>
</tr>
<tr>
<td>Displacement (D)</td>
<td>0</td>
<td>0</td>
<td>-2</td>
<td>2</td>
<td>0</td>
<td>-</td>
</tr>
<tr>
<td>R</td>
<td>{(3, 2), (5, -2)}</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

RD is defined as the discrete density of the frequency of packets with respect to their displacements, i.e., the lateness and earliness from the original position. Let \( S[k] \) denote the set of reorder events in \( R \) with displacement equal to \( k \). That is:

\[ S[k] = \{ r(m, dm) | dm = k \} \]

Let \( |S[k]| \) be the cardinality of set \( S[k] \). Thus, \( RD[k] \) is defined as \( |S[k]| \) normalized with respect to the total number of received packets \( (N') \). Note that \( N' \) does not include duplicate or lost packets.

\[ RD[k] = |S[k]| / N' \text{ for } k \neq 0 \]

\( RD[0] \) corresponds to the packets for which RI is the same as the sequence number:

\[ RD[0] = 1 - \text{sum}(|S[k]| / N') \]

As defined previously, \( FD[k] \) is the measure that keeps track of \( |S[k]| \).

5. Selection of DT

Although assigning a threshold for determining lost and duplicate packets might appear to introduce error into the reorder metrics, in practice this need not be the case. Applications, protocols, and the network itself operate within finite resource constraints that introduce practical limits beyond which the choice of certain values becomes irrelevant. If the operational nature of an application is such that a DT can be defined, then using DT in the computation of reorder metrics will not invalidate nor limit the effectiveness of...
the metrics, i.e., increasing DT does not provide any benefit. In the case of TCP, the maximum transmit and receive window sizes impose a natural limit on the useful value of DT. Sequence number wraparound may provide a useful upper bound for DT in some instances.

If there are no operational constraints imposed by factors as described above, or if one is purely interested in a more complete picture of reordering, then DT can be made as large as required. If DT is equal to the length of the packet sequence (worst case scenario), a complete picture of reordering is seen. Any metric that does not rely on a threshold to declare a packet as lost implicitly makes one of two assumptions: a) A missing packet is not considered lost until the end of the sequence, or b) the packet is considered lost until it arrives. The former corresponds to the case where DT is set to the length of the sequence. The latter leads to many problems related to complexity and robustness.

6. Detection of Lost and Duplicate Packets

In RD, a packet is considered lost if it is late beyond DT. Non-duplicate arriving packets do not have a copy in the buffer and do not have a sequence number less (earlier) than E. In RBD, a packet is considered lost if the buffer is filled to its threshold BT. A packet is considered a duplicate when the sequence number is less than the expected packet, or if the sequence number is already in the buffer.

Since RI skips the sequence number of a lost packet, the question arises as to how to assign an RI to subsequent packets that arrive before it is known that the packet is lost. This problem arises only when reorder metrics are calculated in real-time for an incoming sequence, and not with offline computations. This concern can be handled in one of two ways:

a) Go-back Method: RD is computed as packets arrive. When a packet is deemed lost, RI values are corrected and displacements are recomputed. The Go-back Method is only invoked when a packet is lost and recomputing RD involves at most DT packets.

b) Stay-back Method: RD evaluation lags the arriving packets so that the correct RI and E values can be assigned to each packet as it arrives. Here, RI is assigned to a packet only once, and the value assigned is guaranteed to be correct. In the worst case, the computation lags the arriving packet by DT. The lag associated with the Stay-back Method is incurred only when a packet is missing.
Another issue related to a metric and its implementation is the robustness against peculiarities that may occur in a sequence as discussed in Section 2. Consider, for example, the arrival sequence (1, 5430, 2, 3, 4, 5, ...). With RD, a sense of proportionality is easily maintained using the concept of threshold (DT), which limits the effects a rogue packet can have on the measurement results. In this example, when the displacement is greater than DT, rogue packet 5430 is discarded. In this way the impact due to the rogue packet is limited, at most, to DT packets, thus imposing a limit on the amount of error it can cause in the results. Note also that a threshold different from DT can be used for the same purpose. For example, a pre-specified threshold that limits the time a packet remains in the buffer can make RBD robust against rogue packets.

7. Algorithms to Evaluate RD and RBD

The algorithms to compute RD and RBD are given below. These algorithms are applicable for online computation of an incoming packet stream and provide an up-to-date metric for the packet stream read so far. For simplicity, the sequence numbers are considered to start from 1 and continue in increments of 1. Only the Stay-back Method of loss detection is presented here; hence, the RD values lag by a maximum of DT. The algorithm for the Go-back Method is given in [Bar04]. Perl scripts for these algorithms are posted in [Per04].

7.1. Algorithm for RD

Variables used:

<table>
<thead>
<tr>
<th>Variable</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>RI</td>
<td>receive index</td>
</tr>
<tr>
<td>S</td>
<td>Arrival under consideration for lateness/earliness computation</td>
</tr>
<tr>
<td>D</td>
<td>Lateness or earliness of the packet being processed: dm for m.</td>
</tr>
<tr>
<td>FD[-DT..DT]</td>
<td>Frequency of lateness and earliness.</td>
</tr>
<tr>
<td>window[1..DT+1]</td>
<td>List of incoming sequence numbers; FIFO buffer.</td>
</tr>
<tr>
<td>buffer[1..DT]</td>
<td>Array to hold sequence numbers of early arrivals.</td>
</tr>
</tbody>
</table>

Step 1. Initialize:

Store first unique DT+1 sequence numbers in arriving order into window; RI = 1;

Step 2. Repeat (until window is empty):

If (window or buffer contains sequence number RI)
{ Move sequence number out of window to S # window is FIFO
D = RI - S; # compute displacement

If (absolute(D) <= DT) # Apply threshold
{
    FD[D]++; # Update frequency
    If (buffer contains sequence number RI)
        Delete RI from buffer;
    If (D < 0) # Early Arrival
        add S to empty slot in buffer;
        RI++; # Update RI value
}
Else # Displacement beyond threshold.
{
    Discard S;
    # Note, an early arrival in window is moved to buffer if
    # its displacement is less or equal to DT. Therefore, the
    # contents in buffer will have only possible RIs. Thus,
    # clearing an RI as it is consumed prevents memory leaks
    # in buffer
}
# Get next incoming non-duplicate sequence number, if any.
newS = get_next_arrival(); # subroutine called
if (newS != null)
{
    add newS to window;
}
if (window is empty) go to step 3;
Else # RI not found. Get next RI value.
{
    # Next RI is the minimum among window and buffer contents.
    m = minimum (minimum (window), minimum (buffer));
    If (RI < m)
        RI = m;
    Else
        RI++;
}

Step 3. Normalize FD to get RD;

# Get a new sequence number from packet stream, if any
subroutine get_next_arrival()
{
    do # get non-duplicate next arrival

newS = new sequence from arriving stream;
if (newS == null) # End of packet stream
    return null;
while (newS < RI or newS in buffer or newS in window);
    return newS;
}

7.2. Algorithm for RBD

Variables used:

# E : Next expected sequence number.
# S : Sequence number of the packet just arrived.
# B : Current buffer occupancy.
# BT: Buffer Occupancy threshold.
# FB[i]: Frequency of buffer occupancy i (0 <= i <= BT).
# in_buffer(N) : True if the packet with sequence number N is already stored in the buffer.

=================================================================
1. Initialize E = 1, B = 0 and FB[i] = 0 for all values of i.
2. Do the following for each arrived packet.

    If (in_buffer(S) || S < E) /*Do nothing*/;
    /* Case a: S is a duplicate or excessively delayed packet.
    Discard the packet.*/
    Else
    {
        If (S == E)
        /* Case b: Expected packet has arrived.*/
        {
            E = E + 1;
            While (in_buffer(E))
            {
                B = B - 1; /* Free buffer occupied by E.*/
                E = E + 1; /* Expect next packet.*/
            }
            FB[B] = FB[B] + 1; /*Update frequency for buffer occupancy B.*/
        } /* End of If (S == E)*/
        ElseIf (S > E)
        /* Case c: Arrived packet has a sequence number higher than expected.*/
        {
            /* Code */
        }
    }
    /* End of If (in_buffer(S) || S < E) */
If (B < BT)
/* Store the arrived packet in a buffer.*/
B = B + 1;
Else
/* Expected packet is delayed beyond the BT.
Treat it as lost.*/
{
    Repeat
    {
        E = E + 1;
    }
    Until (in_buffer(E)  ||  E == S);

    While (in_buffer(E)  ||  E == S)
    {
        if (E != S) B = B - 1;
        E = E + 1;
    }
    FB[B] = FB[B] + 1; /*Update frequency for buffer
occupancy B.*/
} /* End of ElseIf (S > E)*/

3. Normalize FB[i] to obtain RBD[i], for all values of i using

\[
RBD[i] = \frac{FB[i]}{\sum(FB[j] \text{ for } 0 \leq j \leq BT)}
\]

8. Examples

a. Scenario with no packet loss

Consider the sequence of packets (1, 4, 2, 5, 3, 6, 7, 8) with DT = BT = 4.

Tables 1 and 2 show the computational steps when the RD algorithm is
applied to the above sequence.
Table 1: Late/Early-packet Frequency computation steps

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>4</th>
<th>2</th>
<th>5</th>
<th>3</th>
<th>6</th>
<th>7</th>
<th>8</th>
</tr>
</thead>
<tbody>
<tr>
<td>RI</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>5</td>
<td>6</td>
<td>7</td>
<td>8</td>
</tr>
<tr>
<td>D</td>
<td>0</td>
<td>-2</td>
<td>1</td>
<td>-1</td>
<td>2</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>FD[D]</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>4</td>
</tr>
</tbody>
</table>

(S, RI, D and FD[D] as described in Section 7.1)

The last row (FD[D]) represents the current frequency of occurrence of the displacement D, e.g., column 3 indicates FD[1] = 1 while column 4 indicates FD[-1] = 1. The final set of values for RD are shown in Table 2.

Table 2: Reorder Density (RD)

<table>
<thead>
<tr>
<th>D</th>
<th>-2</th>
<th>-1</th>
<th>0</th>
<th>1</th>
<th>2</th>
</tr>
</thead>
<tbody>
<tr>
<td>FD[D]</td>
<td>1</td>
<td>1</td>
<td>4</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>RD[D]</td>
<td>0.125</td>
<td>0.125</td>
<td>0.5</td>
<td>0.125</td>
<td>0.125</td>
</tr>
</tbody>
</table>

(D, FD[D] and RD[D] as described in Section 7.1)

Tables 3 and 4 illustrate the computational steps for RBD for the same example.

Table 3: Buffer occupancy frequencies (FB) computation steps

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>4</th>
<th>2</th>
<th>5</th>
<th>3</th>
<th>6</th>
<th>7</th>
<th>8</th>
</tr>
</thead>
<tbody>
<tr>
<td>E</td>
<td>1</td>
<td>2</td>
<td>2</td>
<td>3</td>
<td>3</td>
<td>6</td>
<td>7</td>
<td>8</td>
</tr>
<tr>
<td>B</td>
<td>0</td>
<td>1</td>
<td>1</td>
<td>2</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>FB[B]</td>
<td>1</td>
<td>1</td>
<td>2</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>5</td>
</tr>
</tbody>
</table>

(E, S, B and FB[B] as described in Section 7.2)
Table 4: Reorder Buffer-occupancy Density

<table>
<thead>
<tr>
<th>B</th>
<th>0</th>
<th>1</th>
<th>2</th>
</tr>
</thead>
<tbody>
<tr>
<td>FB[B]</td>
<td>5</td>
<td>2</td>
<td>1</td>
</tr>
<tr>
<td>RBD[B]</td>
<td>0.625</td>
<td>0.25</td>
<td>0.125</td>
</tr>
</tbody>
</table>

(B, FB[B] and RBD[B] as discussed in Section 7.2)

Graphical representations of the densities are as follows:

```
  ^                            ^
  |                            |
  ^                            |
  |                            | 0.5 | 0.625
  |                            |
RD[D] | _ | _ | _ | _ | _ | _ | 0.125 | RBD[B] | | | | | - 0.25
| | | | | | | | | | | | | | - 0.125
+----------->                  ---+----------+
-2 -1 0 1 2 2 0 1 2 2 2 2 1 0 B -->
```

b. Scenario with packet loss

Consider a sequence of 6 packets (1, 2, 4, 5, 6, 7) with DT = BT = 3. Table 5 shows the computational steps when the RD algorithm is applied to the above sequence to obtain FD[D].

Table 5: Late/Early-packet Frequency computation steps

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>2</th>
<th>4</th>
<th>5</th>
<th>6</th>
<th>7</th>
</tr>
</thead>
<tbody>
<tr>
<td>RI</td>
<td>1</td>
<td>2</td>
<td>4</td>
<td>5</td>
<td>6</td>
<td>7</td>
</tr>
<tr>
<td>D</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>FD[D]</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>5</td>
<td>6</td>
</tr>
</tbody>
</table>

(S, RI, D and FD[D] as described in Section 7.1)
Table 6 illustrates the FB[B] for the above arrival sequence.

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>2</th>
<th>4</th>
<th>5</th>
<th>6</th>
<th>7</th>
</tr>
</thead>
<tbody>
<tr>
<td>E</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>3</td>
<td>3</td>
<td>7</td>
</tr>
<tr>
<td>B</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>0</td>
</tr>
<tr>
<td>FB[B]</td>
<td>1</td>
<td>2</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>3</td>
</tr>
</tbody>
</table>

(E, S, B and FB[B] as described in Section 7.2)

Graphical representations of RD and RBD for the above sequence are as follows.

```
  ^                        ^
  |                        |
^ |                        | 0.5 |
  |                        |     |
^ | |                      | |
  | |                ^      |
  | |    1.0   _          |
RD[D] | |               RBD[B] | |
      | |                       |
      | |                      ||
      | |                     || |
      | |                    ||  |
      | |                  0.167
```

(c. Scenario with duplicate packets)

Consider a sequence of 6 packets (1, 3, 2, 3, 4, 5) with DT = 2. Table 7 shows the computational steps when the RD algorithm is applied to the above sequence to obtain FD[D].

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>3</th>
<th>2</th>
<th>3</th>
<th>4</th>
<th>5</th>
</tr>
</thead>
<tbody>
<tr>
<td>RI</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>-</td>
<td>4</td>
<td>5</td>
</tr>
<tr>
<td>D</td>
<td>0</td>
<td>-1</td>
<td>1</td>
<td>-</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>FD[D]</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>-</td>
<td>2</td>
<td>3</td>
</tr>
</tbody>
</table>

(S, RI, D and FD[D] as described in Section 7.1)
Table 8 illustrates the FB[B] for the above arrival sequence.

<table>
<thead>
<tr>
<th>S</th>
<th>1</th>
<th>3</th>
<th>2</th>
<th>3</th>
<th>4</th>
<th>5</th>
</tr>
</thead>
<tbody>
<tr>
<td>E</td>
<td>1</td>
<td>2</td>
<td>2</td>
<td>-</td>
<td>4</td>
<td>5</td>
</tr>
<tr>
<td>B</td>
<td>0</td>
<td>1</td>
<td>0</td>
<td>-</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>FB[B]</td>
<td>1</td>
<td>1</td>
<td>2</td>
<td>-</td>
<td>3</td>
<td>4</td>
</tr>
</tbody>
</table>

(E, S, B and FB[B] as described in Section 7.2)

Graphical representations of RD and RBD for the above sequence are as follows:

9. Characteristics Derivable from RD and RBD

Additional information may be extracted from RD and RBD depending on the specific applications. For example, in the case of resource allocation at a node to recover from reordering, the mean and variance of buffer occupancy can be derived from RBD. For example:

Mean occupancy of recovery buffer = \( \sum (i \times \text{RBD}[i]) \) for \( 0 \leq i \leq \text{BT} \)

The basic definition of RBD may be modified to count the buffer occupancy in bytes as opposed to packets when the actual buffer space is more important. Another alternative is to use time to update the buffer occupancy compared to updating it at every arrival instant.

The parameters that can be extracted from RD include the percentage of late (or early) packets, mean displacement of packets, and mean displacement of late (or early) packets [Ye06]. For example, the fraction of packets that arrive after three or more of their successors according to the order of transmission is given by Sum

[RD[i] for 3<=i<=DT]. RD also allows for extraction of parameters such as entropy of the reordered sequence, a measure of disorder in the sequence [Ye06]. Due to the probability mass function nature of RD, it is also a convenient measure for theoretical modeling and analysis of reordering, e.g., see [P106].

10. Comparison with Other Metrics

RD and RBD are compared to other metrics of [RFC4737] in [P107].

11. Security Considerations

The security considerations listed in [RFC4737], [RFC3763], and [RFC4656] are extensive and directly applicable to the usage of these metrics; thus, they should be consulted for additional details.

12. References

12.1. Normative References


12.2. Informative References


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