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The Quality for Service Protocol
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Abstract

This memo describes an application level protocol for the standard communication of e2e QoS compliance information using a protocol based on Hypertext Transfer Protocol (HTTP), which forms the basis for the World Wide Web, and Session Description Protocol (SDP). Quality HTTP (Q4S) provides a mechanism for latency, jitter, bandwidth and packet loss negotiation and monitoring, alerting whenever one of the negotiated conditions is violated.

Implementation details on the actions to be triggered upon reception/detection of QoS alerts exchanged by the protocol are out of scope of this draft, it is application dependant (e.g. increase quality, reduce bit-rate) or even network dependant (e.g. change connection's quality profile).

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1. Introduction

The World Wide Web (WWW) is a distributed hypermedia system which has gained widespread acceptance among Internet users. Although WWW browsers support other, preexisting Internet application protocols, the native and primary protocol used between WWW clients and servers is the HyperText Transfer Protocol (HTTP) (RFC 2616 [1]). The ease of use of the Web has prompted its widespread employment as a client/server architecture for many applications. Many of such applications require the client and the server to be able to communicate each other and exchange information with certain quality constraints.

Quality in communications at application level consists of four measurable parameters:

- o Latency: The time a message takes to travel from source to destination. It may be approximated to $RTT/2$ (Round trip time), assuming the networks are symmetrical.
- o Jitter: latency variation. There are some formulas to calculate Jitter, and in this context we will consider the statistical variance formula.
- o Bandwidth: To assure the quality, a protocol MUST assure the availability of bandwidth needed by the application.
- o Packet loss: The percentage of packet loss is closely related to bandwidth and jitter. Affects bandwidth because a high packet loss implies sometimes retransmissions that also consumes extra bandwidth, other times the retransmissions are not achieved (for example in video streaming over UDP) and the information received is less than the required bandwidth. In terms of jitter, a packet loss sometimes is seen by the destination like a larger time between arrivals, causing a jitter growth.

Q4S provides a mechanism for quality monitoring and it is based on HTTP and SDP in order to be easily integrated in WWW, but it may be used by any type of application, not only those based on HTTP. Quality requirements may be needed by any type of application that communicates using any kind of protocol, especially those which have real-time constraints.

Q4S is an application level Client/Server protocol which tries to measure continuously session quality for a given flow (or set of flows), end-to-end and in real-time; raising an alert if quality parameters are below a given threshold. The thresholds of each application are different, depending on the nature of each application. Q4S does not describe either the actions carried out to deal with the alert or how to implement them.

Q4S is session-independent from the application flows, in order to minimize the impact on them. To perform the measurements, two control flows are created on either direction (forward and reverse).

1.1. Motivation

Monitoring quality of service (QoS) in computer networks is useful for several reasons:

- o Enable real-time services and applications to verify whether network resources achieve a certain QoS level.

- o Monitoring helps real-time services and applications to run through the Internet, allowing the existence of Application Content providers (ACPs) which offer guaranteed real-time services to the final users.
- o Monitoring also applies to Peer to Peer (P2P) real-time applications
- o Enable ISPs to offer QoS to any ACP or final user application in an accountable way
- o Enable e2e negotiation of QoS parameters, from any ISP to any ISP.

A protocol to monitor QoS must address the following issues:

- o Must be ready to be used in conjunction with current standard protocols and applications, without forcing a change on them.
- o Must have a formal and compact way to specify quality constraints of the desired application to run.
- o Must have measurement mechanisms avoiding application disruption.
- o Must have specific messages to alert about the violation of quality constraints in different directions (forward and reverse), because network routing may not be symmetrical, and of course, quality constraints may not be symmetrical.
- o Must Protect the data (constrains, measurements, QoS levels asked to the network) in order to avoid the injection of malicious data in the measurements.

1.2. Summary of Features

Quality for Service is a message-oriented communication protocol that can be used in conjunction with any other application-level protocol.

The benefits in quality enhancement provided by Q4S can be used by any type of application that uses any type of protocol for data transport. It provides a quality monitoring scheme to any communication that takes place between the client and the server, not only the Q4S communication itself.

Q4S does not establish multimedia sessions and it does not transport application data. The type of use and kind of protocol of this quality communication is application dependant and can be whatever. Q4S doesn't force any particular protocol or way of using of the quality connection.

Q4S session lifetime is composed of four phases with different purposes, and inside each phase a negotiated measurement procedure is used. Different measurement procedures can be used inside Q4S (although for compatibility reasons a default measurement mechanism is defined). Basically, Q4S only defines how to transport SLA information and measurement results as well as providing some mechanisms for alerting.

Q4S MUST be executed just before starting a client-server application which needs a quality connection in terms of latency, jitter, bandwidth and packet loss. Once client and server have succeeded in establishing communication under quality constraints, the application can start, and Q4S continues measuring and alerting.

During the lifetime of the quality session, the protocol stays in a special state in which it periodically renews the session and alerts if the measurements of quality parameters do not meet the negotiated application requirements.

The quality parameters can be suggested by the client in the first message, but the server can accept these parameter values or force others. The server is in charge of deciding the final values of quality connection.

2. Terminology

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in RFC 2119 [3].

3. Overview of Operation

This section introduces the basic operation of Q4S using simple examples. This section is of tutorial nature and does not contain any normative statements.

3.1. Protocol Phases

All elements of the IP network contribute to the quality in terms of latency, jitter, bandwidth and packet loss. All these elements have their own quality policies in terms of priorities, traffic mode, etc. and each element has its own way to manage the quality. The purpose of a quality connection is to establish an end-to-end communication with enough quality for the application.

To monitor negotiated SLA compliance, four phases are defined

- o Handshake phase: in which the server is contacted by the client and in the answer message the quality constraints for the application is communicated.
- o Negotiation phase: in which the quality of the connection is measured in both directions (latency, jitter, bandwidth and packet loss), and Q4S messages are sent in order to alert when the quality does not match the constraints. This phase is iterative until quality constraints are reached or the session is cancelled after checking that the quality constraints are impossible to reach. Just after reaching the quality requirements, Q4S provides a simple optional mechanism to start the application which will benefit from quality connection, using HTTP.
- o Continuity phase: in which quality is continuously measured. If quality becomes degraded, an alert shall be released. New measurements may follow up to a negotiated maximum before moving to Termination phase. In this phase the measurements MUST avoid disturbing application by consuming network resources.
- o Termination phase: in which the session is terminated.

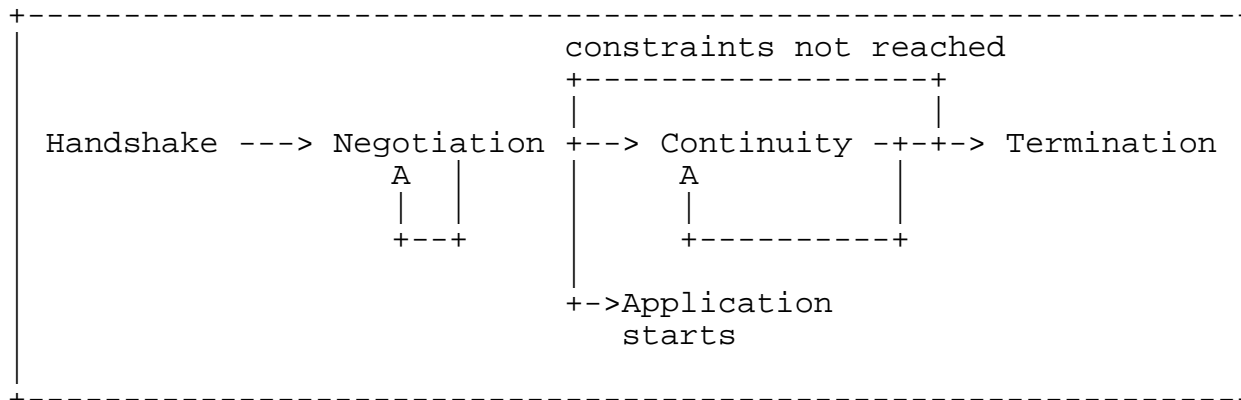


Figure 1 Session lifetime phases.

3.1.1. Handshake Phase

The first phase consists of a Q4S BEGIN method issued from the client to the server.

The first Q4S message MUST have a special URI (RFC 3986 [4]), which forces the use of the Q4S protocol if it is implemented in a standard web browser.

This URI, named "Contact URI", is used to request the start of a session. Its scheme MUST be:

```
"q4s:" "://" host [":" port] [path["?" query]
```

Optionally, the client can send the desired quality parameters (enclosed in the body of the message as an SDP document) and the server can take them into account when it builds the answer with the final values, following an offer / answer schema (RFC 3464 [5]). The description of these quality parameters are attached in an SDP document.

If the request is accepted, the server MUST answer with a Q4S 200 OK message, and in the body of the answer message, an SDP document MUST be included (RFC 4566 [2]), with information about the required quality constraints. Q4S responses should use the protocol designator "Q4S/1.0".

After these two messages are exchanged, the first phase is completed. The quality parameters have been sent to the client. Next

step is to measure the quality of the communication path between the client and the server and alert if the SLA is being violated.

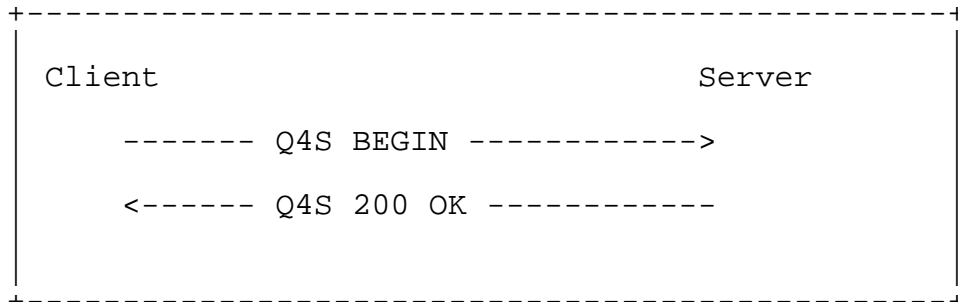


Figure 2 Regular handshake phase.

Example of Client Request and server answer:

Client Request:

```

=====
BEGIN q4s://www.example.com Q4S/1.0
Content-Type: application/sdp
User-Agent: q4s-ua-experimental-1.0
Content-Length: 142

```

(SDP not shown)

Server Answer:

```

=====
Q4S/1.0 200 OK
Date: Mon, 10 Jun 2010 10:00:01 GMT
Content-Type: application/sdp
Expires: 3000
Q4S-Resource-Server:
q4s://www.example.com/example/util/agent?num=666
Q4S-Policy-Server: q4s://www.qosmanager.com/agent
Signature: 6eclba40e2adf2d783de530ae254acd4f3477ac4
Content-Length: 131

```

(SDP not shown)

The "Expires" header purpose is to provide a sanity check and enables the server to close inactive sessions. If the client does

not send a new request before the expiration time, the server can close the session.

The "Signature" header contains a digital signature that can be used by the network to validate the SDP, preventing security attacks.

The signature is an optional header generated by the server using a hash and encryption method such as MD5 (RFC 1321 [6]) and RSA (RFC 2437 [7]), but it depends on the certificate used by the server. This certificate is supposed to be delivered by a Certification Authority (CA) or policy owner to the server. The signature is applied to the SDP body.

```
Signature= RSA ( MD5 (<sdp>), <certificate> )
```

If the signature is not present, other validation mechanism may be implemented in order to provide assured quality with security and control.

The optional response header "Q4S-Resource-Server" contains the Session URI, which is in charge of this session. This URI MUST be invoked by the client in all later requests. Example:

```
Q4S-Resource-server:  
q4s://www.example.com/example/util/agent?num=666
```

If this header is not present, the client will continue sending all requests to the original Contact URI, but if it is present, its use is mandatory.

The last optional response header is "Q4S-Policy-Server" which contains the "Policy Server URI" towards which client MUST send the later QOS-ALERT messages. This header will be explained later on. In case this header is present, the Q4S-Resource-server header is mandatory.

During the next phases of the protocol, the client role will perform a mix of client and server role. Hence, the client can specify a "Q4S-Resource-Client" header in the BEGIN request, indicating the Resource Client URI, a relative URI in charge of the server requests when client receives requests from the server. Example:

```
Q4S-Resource-Client: /example/useragent
```

This URI MUST be relative because user agents may not have an associated domain, or its IP address is unknown.

3.1.1.1. Description of Quality parameters inside SDP

The original goal of SDP was to announce necessary information for the participants and multicast MBONE (Multicast Backbone) applications. Right now, its use has been extended to the announcement and the negotiation of multimedia sessions. The purpose of SDP in the Q4S context is different because no media parameters are set, therefore the number of media attributes ("m") is always zero. This is because Q4S purpose is not to establish media stream sessions, but to monitor a quality connection, and this quality connection can be used to establish media sessions by other protocols, or for any other purpose.

The SDP embedded in the messages is the container of the quality parameters. The included information can comprise all or some of the following parameters, by means of optional session-level attributes:

- o QoS level for uplink and downlink: specified in the "qos-level" attribute. Default values are 0 for both directions. The meaning of each level is out of scope of Q4S, but, in general, a higher level should correspond to a better service quality.
- o Maximum latency tolerance for uplink and downlink: specified in the "latency" attribute, expressed in milliseconds.
- o Maximum jitter tolerance for uplink and downlink: specified in the "jitter" attribute, expressed in milliseconds.
- o Minimum bandwidth for uplink and downlink: specified in the "bandwidth" attribute, expressed in kbps.
- o Maximum packet loss tolerance for uplink and downlink: specified in the "packetloss" attribute expressed in percentage.
- o Flows (protocol, source IP, source Port + destination IP, destination port) of data over TCP and UDP ports to be used in uplink and downlink: specified in the "flow" attribute.
- o Measurement procedure and results of quality measurements: specified in the "measurement" attribute.

This is an example of SDP for Q4S usage. For each attribute two values separated by "/" are involved. These values represent the uplink and downlink values: <uplink> / <downlink>. When one or both

of these values are empty, it means that there is no constraint on this parameter.

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:0/0
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,75/75,,0
a=measurement:latency 10000/10000
a=measurement:jitter 10000/10000
a=measurement:bandwidth 0/0
a=measurement:packetloss 0/0
```

Inside the constraints, several "flow" attributes can be defined. The goal is to monitor each flow to verify that the quality constraints are met. These flows include the type (uplink or downlink), the protocol (TCP or UDP) (RFC 761 [8] and RFC 768 [9]) and the ports that are going to be used by the application data and, of course, by the control (for quality measurements), because the quality measurements MUST be achieved over the same quality session for each direction. All defined flows will be considered within the same quality profile, which is determined by the qos-level attribute in each direction.

During negotiation phase control ports will be used for Q4S messages, and this is the reason to separate application data ports from Q4S control ports, otherwise they could collide.

The control should involve two UDP flows (one for uplink and other for downlink) and two TCP flows (one for uplink and other for downlink), but application data could involve many flows, depending on the nature of the application. The handshake phase takes place through the Contact URI, using TCP port 80 for example. However, the

negotiation phase will take place on the specified control ports (UDP and TCP) using the Session URI.

A "downlink port" is a port in which the client listens for server requests (and MUST be used as origin port of client responses), while an "uplink port" is a port in which server is listening incoming messages from the client (and MUST be used as origin port of server responses).

Server's SDP information on "downlink" ports is informational for the client (it could even be a null value meaning that they could be chosen randomly as per OS standard rules). "Downlink" ports inside the SDP must always be matched against actual received port values on the server side in order deal with NAT/NATP devices. Example:
a=flow:control downlink TCP/0

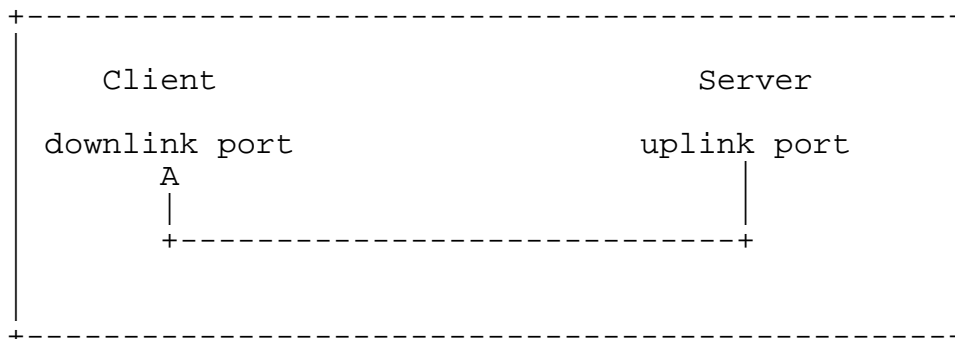


Figure 3 Downlink flow.

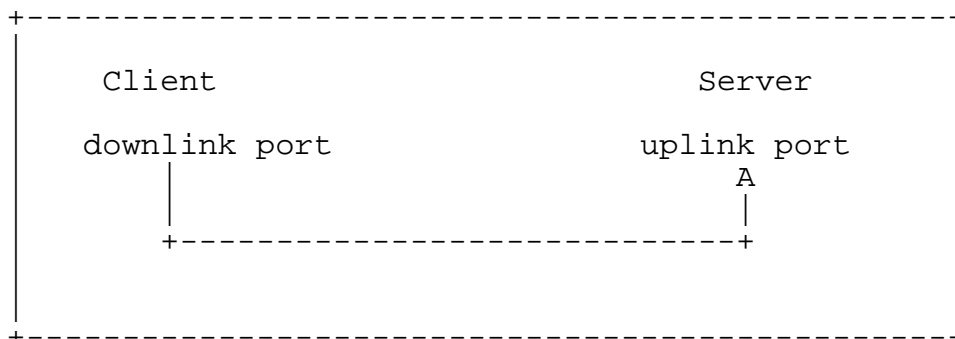


Figure 4 Uplink flow.

In addition, measurement parameters are included using the session attribute "measurement". The first measurement parameter is the procedure. By default, Q4S provides a "default" procedure for measurement, but others like RTP/RTCP might be used. In the initial client request a set of measurement procedures can be sent to the server for negotiation (one line MUST be included in SDP for each one). The server will answer with only one line with the chosen procedure.

For each procedure, a set of values of parameters can be included in the same attribute line, as in the following example:

```
a=measurement:procedure default,50/50,75/75,5000,0
```

Where the procedure name is "default" and one parameter is included separated by ",". The meaning of each value depends on the procedure. In the procedure "default", the meaning of these parameters are:

- o The first parameter is the interval of time (in milliseconds) between PING requests during the negotiation phase. Forward and reverse values from the client's point of view are separated by "/". This allows having different responsiveness values depending on the control resources used in each direction.
- o The second parameter is the interval of time (in milliseconds) between PING requests during the continuity phase. Forward and reverse values are separated by "/". This allows having two different responsiveness values depending on the control resources used in each direction.
- o The third parameter is the time used to measure bandwidth during the negotiation phase. If not present, a default value of 5000 ms will be assumed. Forward and reverse values are separated by "/".
- o The fourth parameter indicates the mode for continuity phase (0 means "normal" and 1 means "sliding window"). If not present, normal mode (default value of 0) will be assumed.

Quality parameters read by the procedure provide a snapshot of the quality level reached in each stage.

Since the handshake phase does not make any measurement, this section could be empty or filled with dummy values, except procedure, which is mandatory to start the next protocol phase.

3.1.2. Quality negotiation phase

This phase depends on the chosen procedure. The following description corresponds to "default" procedure.

The negotiation phase involves iterations of sequences of messages until the quality session is compliant with the minimum quality constraints or until the quality session is terminated due to the impossibility to meet the constraints.

In order to measure the quality parameters, the client and server can use different mechanisms. This document only describes the "default" mechanism, but others can be used, like RTP/RTCP (RFC 3550 [10]). Measurement of latency and jitter is done calculating the differences in arrival times of packets. This measurement can be achieved with little bandwidth consumption, whereas bandwidth measurement involves higher bandwidth consumption in both directions (uplink and downlink).

Therefore the measurements involve two stages:

- o Stage 0: Measurement of latencies, jitters and packet loss
- o Stage 1: Measurement of bandwidths and packet loss

Notice that packet loss can be measured in both parts, because the messages used for measure latencies can also be used for packet loss measurement.

These two parts are executed sequentially in order to save network resources. If the required latencies and jitters can not be reached, it makes no sense to waste network resources measuring bandwidth. In addition, if the achievement of the required latency and jitter implies upgrading the quality session level, the chance of succeeding in bandwidth measurement without retries is higher, saving network traffic.

The client starts the negotiation phase sending a READY request using the TCP control ports defined in the SDP. This READY request includes an "Stage" header that indicates the measurement stage.

The motivation for this READY message is to synchronize negotiation phases in multiple quality sessions (see 4.2) enabling the possibility to repeat a successful stage.

If either jitter or latency are specified, the negotiation phase begins with the measurement of latencies and jitters (stage 0). If none of those attributes are specified, stage 0 is skipped.

3.1.2.1. Stage 0: Measurement of latencies and jitters

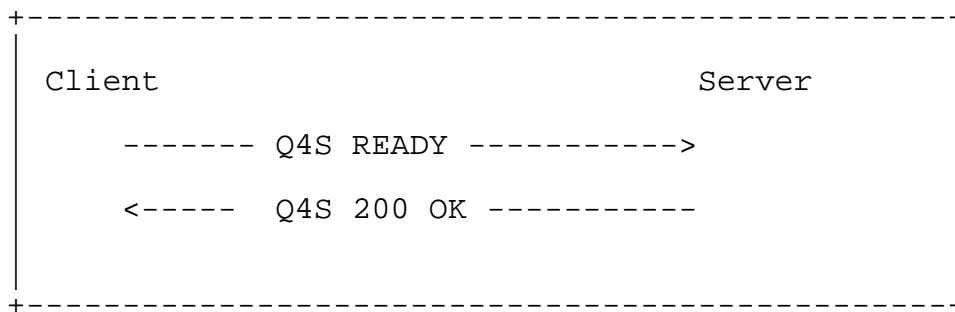


Figure 5 Beginning of the Negotiation phase.

Client Request:

```

=====
READY q4s://www.example.com Q4S/1.0
Stage: 0
Session-Id: 53655765
User-Agent: q4s-ua-experimental-1.0
Content-Length: 0
=====

```

Server Response:

```

=====
Q4S/1.0 200 OK
Session-Id: 53655765
Stage:0
Content-Length: 0
=====

```

Following this, the client and the server start exchanging a number of PING requests and responses that will lead to the calculation of RTT, jitter and packet loss.

The server MUST send its PING requests using the UDP control flow ports defined in the SDP negotiated during the handshake phase. The downlink port is set as destination and the uplink port is set as origin (according to the example, from client UDP port 56000 to server UDP port 55000).

At the same time the server must begin to do exactly the same, using UDP control ports to send PING requests towards the clients.

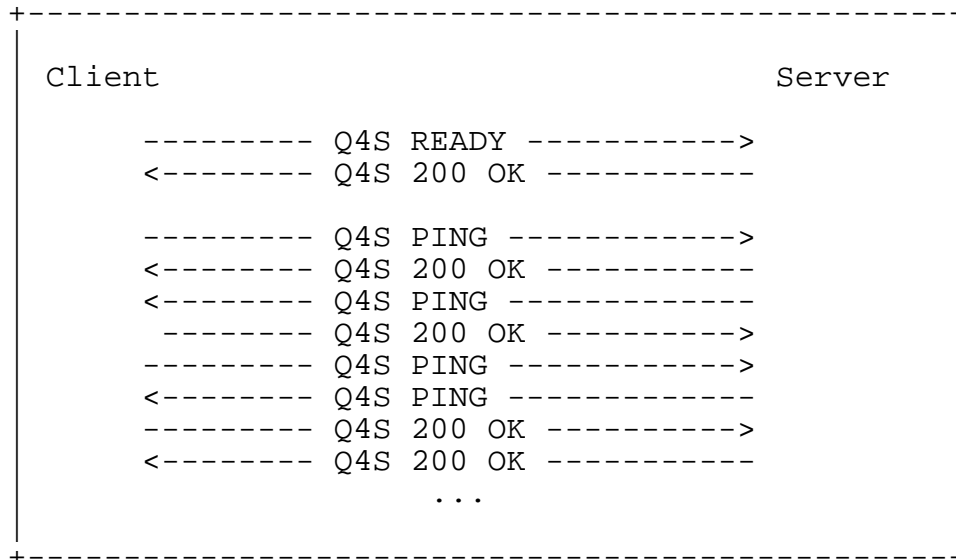


Figure 6 Simultaneous exchange of PING request and responses.

This is an example of the message sent from the client and the server response:

Client Request:

```

=====
PING q4s://www.example.com Q4S/1.0
Session-Id: 53655765
Sequence-Number: 0
User-Agent: q4s-ua-experimental-1.0
Content-Length: 0
=====

```

Server Response:

```

=====
Q4S/1.0 200 OK
Session-Id: 53655765
Sequence-Number: 0
Content-Length: 0
=====

```

The meaning of this method is similar to the ICMP echo request message. Basically the server MUST answer as soon as it receives the message.

Both endpoints MUST send Q4S PING messages periodically, using the same UDP control ports always and do not need to wait for a response to send the next PING request. They just sends PING messages periodically with a different "Sequence-Number" header value. This value is a sequential integer number and MUST start at zero. If this stage is repeated, the initial Sequence-Number MUST start again at zero.

Optionally the PING request can include a "Timestamp" header, with the time in which the message has been sent. In case the header is present, the server MUST include the header in the response without changing the value.

During this phase, the interval between PING requests is defined in the first parameter of the attribute line of SDP where the procedure is specified. In the example, this value is 50 milliseconds (from the client to the server) and 60ms (from the server to the client).

```
a=measurement:procedure default,50/60,50/50,5000,0
```

A couple of correlated messages (request and response matching the Sequence-Number) allow calculating each sample of RTT.

3.1.2.1.1. Round Trip Time calculation

Based on the PING exchange the client is able to calculate the RTT. The RTT is the sum of downlink latency (normally named "reverse latency") and uplink latency (normally named "forward latency").

This process could take a few seconds, and after this time, at least 100 samples of RTT MUST be taken by the client.

3.1.2.1.2. Jitter calculation

Two jitter values are calculated during this stage: uplink jitter and downlink jitter. Downlink jitter is calculated by the client taking into account the PING request messages received from the server. In the same way, uplink jitter is calculated taking into account the PIG request messages received by the server. Note that PING responses are not taken into account when calculating jitter values.

Every time a request message is received by an endpoint (either server or client), the corresponding jitter value is updated using the Statistical Jitter value which is calculated on the first 100 packets received using the statistical variance formula:

$$\text{Jitter Statistical} = \text{SquareRootOf}(\text{SumOf}((\text{ElapsedTime}[i] - \text{Average})^2) / (\text{ReceivedPacketCount} - 1))$$

Hence an endpoint sends a PING periodically with a fixed interval, each value of "elapsed time" (ET) should be very close to this interval. If a PING message is lost, the elapsed time value is doubled, however, this is not an issue because all PING messages are labeled with a Sequence-Number header. Therefore the receiver can discard this elapsed time value. In order to have the first jitter sample, the receiver needs to have 3 PING requests, because each ET is the time between two PINGs and a Jitter needs at least two ET.

After 100 samples the client has the values of RTT and downlink jitter and the server has RTT and uplink jitter.

3.1.2.1.3. Packet loss calculation

Packet loss is measured in both directions. Because Sequence-Number headers are incremented sequentially, the client knows exactly the number of messages lost from the server to the client, and the server knows the number of packets lost from the client to the server.

3.1.2.1.4. Communication of results

After having calculated RTT, jitters and packet loss, the client MUST send a GET request to the server using TCP control port requesting instructions. This message MUST always be sent, independently of the used measurement procedure being used. An SDP carrying the updated values of latency, jitter and packet loss is attached to the body of the request.

As the forward and reverse latencies are unknown, the calculation will assume that the network is symmetric and will assign RTT/2 for uplink and downlink latencies.

Client Request:

```
=====
GET q4s://www.example.com Q4S/1.0
Host: www.example.com
User-Agent: q4s-ua-experimental-1.0
Content-Type: application/sdp
Content-Length: 142

v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:0/0
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,75/75,5000,0
a=measurement:latency 40/40
a=measurement:jitter 0/10
a=measurement:bandwidth 0/0
a=measurement:packetloss 0/2
=====
```

When the server receives this message, it compares the latency value (RTT/2) with its own measurements, in order to avoid inconsistencies.

At this point there are two possibilities

- o The latency, jitter and packet loss constraints are not reached
- o The latency, jitter and packet loss constraints are reached

3.1.2.1.5. Constraints not reached

If the measurements do not meet the quality constraints, the server answers with a 412 message (a precondition setting required by the client or server has not been met).

Server Answer:

```
=====
Q4S/1.0 412 latency
Date: Mon, 10 Jun 2010 10:00:01 GMT
Content-Type: application/sdp
Expires: 3000
Cause: downlink_latency
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Length: 131
```

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:1/0
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,75/75,5000,0
a=measurement:latency 40/40
a=measurement:jitter 20/10
a=measurement:bandwidth 0/0
a=measurement:packetloss 1/2
=====
```

In the 412 message, the server may include a different value for "qos-level" SDP session-level attribute, and the measurements done by the client. All these information MUST be protected using the signature header.

After a 412 message is received by the client, a QOS-ALERT request is sent by the client to acknowledge the SLA violation (using TCP control port). The QOS-ALET request does not have to be answered.

Notice that the server signature header is present in the client request, in order to allow an optional integrity validation.

If the "Q4S-Policy-Server" header was included in the server response of the handshake phase, this message MUST be sent to the URI indicated in that header, otherwise the QOS-ALERT request MUST be sent to the server. This request does not need to be answered.

Client Request:

```
=====
QOS-ALERT q4s://www.example.com Q4S/1.0
Host: www.example.com
User-Agent: q4s-ua-experimental-1.0
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Type: application/sdp
Content-Length: 142
```

(SDP not shown)

```
=====
```

Upon receiving the QOS-ALERT request from the client, the server will issue a QOS-ALERT request towards the client.

Server Request:

```
=====
QOS-ALERT q4s://www.example.com Q4S/1.0
Date: Mon, 10 Jun 2010 10:00:01 GMT
Content-Type: application/sdp
Expires: 3000
Cause: latency
Guard-time: 5000
Signature: 6eclba40e2adf2d783de530ae254acd4f3477ac4
Content-Length: 131
```

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:1/0
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,75/75,5000,0
a=measurement:latency 40/40
a=measurement:jitter 20/10
a=measurement:bandwidth 0/0
a=measurement:packetloss 1/2
=====
```

After the client receives this request, it waits for a while indicated in the server "Guard-time" header, for example to allow different actions to be carried out by the server. (5 seconds should be enough, but this depends on each case) and begins again the measurement process, starting from the beginning (sending the READY request). The maximum qos-level is 9/9 and if this value is reached without reaching the constraints, the quality session is aborted using the CANCEL method, which is detailed further.

If the client does not obeys the "Guard-time", sending the READY message quickly, then the server MUST wait and not answer the READY message until the guard time has elapsed.

If during the measurement process some interference disturbs or affect the measurement results, it is better to repeat the process again rather than alerting of an SLA violation. This is always possible by sending current values of parameter "qos-level" without changes, and in this case a header Guard-time can be set to "0". It is a good practice to repeat the measurements before reporting a violation.

3.1.2.1.6. Constraints not reached with Policy server involved

If during handshake phase the optional header Q4S-Policy-Server is included in the server response, the QOS-ALERT request MUST be sent to the policy server, which should implement all or some of these features (but not exclusive to):

- o Client and server validation in terms of SLA.
- o Authentication (Signature validation) and security (block malicious clients)
- o Policy rules (following rules are only examples):
 - Maximum quality level allowed for the ACP
 - Time bands allowed for provide quality sessions for the ACP
 - Number of simultaneous quality sessions allowed
 - Maximum time used by quality sessions allowed
 - Etc.

With policy server, the QOS-ALERT message sent by the client MUST contain the URIs of the server and the client to be contacted later by the policy server. Therefore the following headers MUST be included in the client request: "Q4S-Resource-server" and "Q4S-Resource-client"

Depending on the results of the operations achieved by the policy server, the client could receive different types of errors or CANCEL messages.

The flows of messages in this case are in the following figure:

It means that the client and the server are ready for bandwidth and packet loss measurement (stage 1).

If the bandwidth constraints are not empty, the negotiation phase continues with stage 1. Otherwise that stage is skipped.

```

+-----+
| Client                                     Server |
|                                             |
| ----- Q4S READY ----->              |
| <----- Q4S 200 OK -----              |
|                                             |
| ----- Q4S PING ----->               |
| <----- Q4S 200 OK -----              |
| <----- Q4S PING -----              |
| ----- Q4S 200 OK ----->            |
| ----- Q4S PING ----->              |
| <----- Q4S PING -----              |
| <----- Q4S 200 OK -----              |
| ----- Q4S 200 OK ----->            |
|                                             |
|           ...                             |
| ----- Q4S GET ----->                |
| <----- Q4S 412 -----                 |
| ----- Q4S QOS-ALERT ----->         |
| <----- Q4S QOS-ALERT -----         |
|           (delay)                         |
| ----- Q4S PING ----->              |
| <----- Q4S PING -----              |
| <----- Q4S 200 OK -----              |
| ----- Q4S 200 OK ----->            |
|                                             |
|           ...                             |
| ----- Q4S GET ----->                |
| <----- Q4S 412 -----                 |
| ----- Q4S QOS-ALERT ----->         |
| <----- Q4S QOS-ALERT -----         |
|           (delay)                         |
| ----- Q4S PING ----->              |
| <----- Q4S PING -----              |
| <----- Q4S 200 OK -----              |
| ----- Q4S 200 OK ----->            |
|                                             |
|           ...                             |
| ----- Q4S GET ----->                |
| <----- Q4S 200 OK -----              |
|
+-----+

```

Figure 8 Latency and jitter measurements with final success

3.1.2.2. Stage 1: Measurement of bandwidth and packet loss

This stage begins in the same way as the previous one, sending a READY request over TCP control ports. This READY message "Stage" header value is 1.

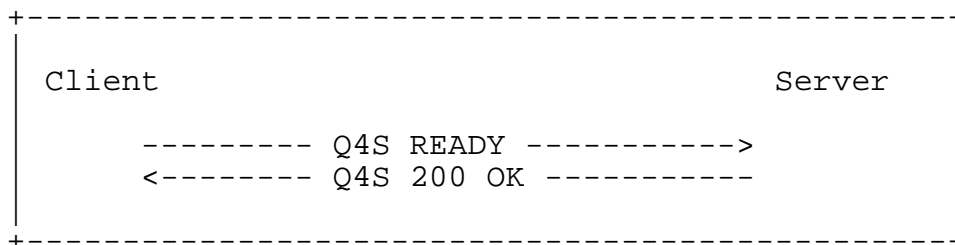


Figure 9 Starting bandwidth and packet loss measurement

Client Request:

```

=====
READY q4s://www.example.com Q4S/1.0
User-Agent: q4s-ua-experimental-1.0
Stage: 1
Session-Id: 53655765
Content-Length: 0
=====

```

Server Response:

```

=====
Q4S/1.0 200 OK
Session-Id: 53655765
Stage: 1
Content-Length: 0
=====

```

Just after receiving the 200 OK, both the client and the server MUST start sending messages simultaneously using the UDP control ports, at the needed rate to reach the bandwidth constraint in each direction using messages of 1 Kbyte length. The messages are sent during a period of time defined in the SDP. This time is the third parameter of procedure "default", in milliseconds. If this parameter is not present, a value of 5 seconds will be used by default.

```
a=measurement:procedure default,50/50,75/75,5000,0
```

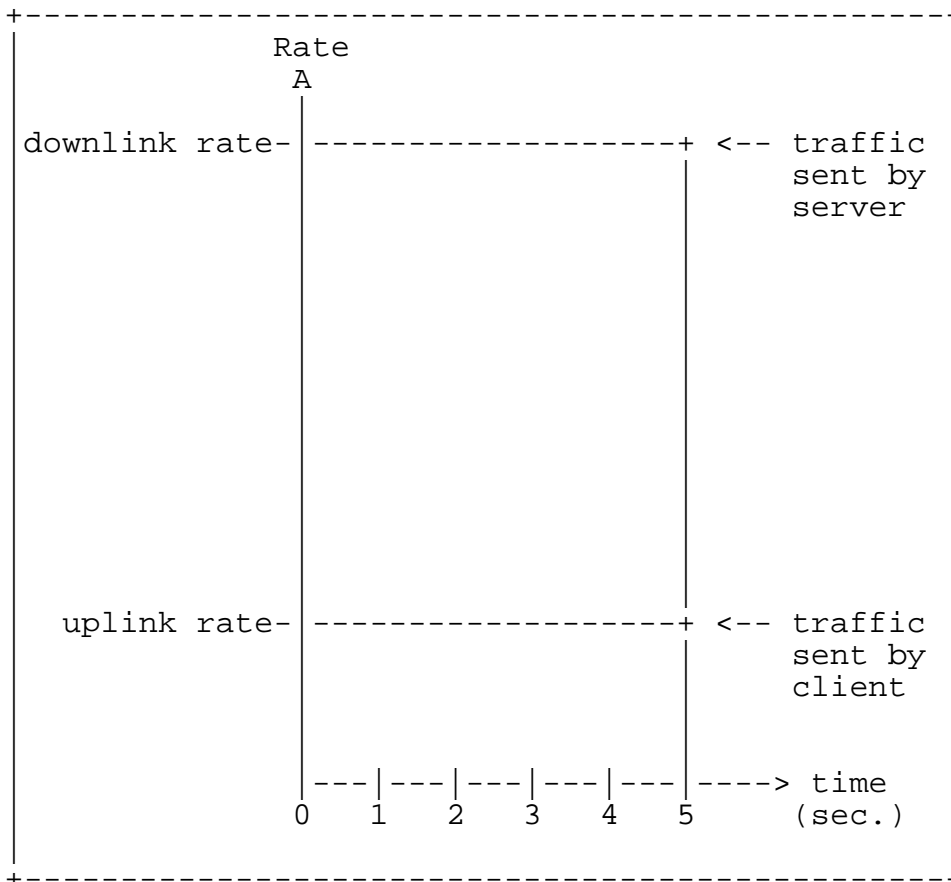


Figure 10 Bandwidth and packet loss measurements.

The goal of this phase is not to measure the internet connection bandwidth connection but to determine if the quality constraints can be reached or not. This is the reason for not sending more bit rate than needed.

All requests to be sent MUST be 1 kilobyte length (UDP payload length), and include a Sequence-Number header with a sequential number starting at 0. If the stage is repeated, the values MUST start again at zero. Examples:

Client message:

=====

```
DATA q4s://www.example.com Q4S/1.0
User-Agent: q4s-ua-experimental-1.0
Session-Id: 53655765
Sequence-Number: 0
Content-Type: text
Content-Length: XXXX
```

```
aaaaaaaaaaaa ( to complete 1024 bytes UDP payload length)
```

=====

The requests MUST NOT be answered, but only sent. The client will send packets to the server in order to allow server measure client bandwidth, and the server will do the same towards the client. The packets have a Sequence-Number to be aware of the packet loss at reception. The value of Sequence-Number will start at zero and will be incremented by 1 for each message.

server message:

=====

```
DATA q4s://www.example.com Q4S/1.0
Session-Id: 53655765
Sequence-Number: 0
Content-Type: text
Content-Length: 1024
```

```
aaaaaaaaaaaa ( to complete 1024 bytes UDP payload length)
```

=====

After a 5 seconds measurements the client has a collection of server messages and may calculate the packet loss and downlink bandwidth received. At the other side, the server has the uplink bandwidth and packet loss.

Client MUST send a GET message to the server using the TCP control port including the SDP data filled up with the measured downlink bandwidth and packet loss.

Client Request:

=====

```
GET q4s://www.example.com Q4S/1.0
Host: www.example.com
User-Agent: q4s-ua-experimental-1.0
Session-Id: 53655765
Content-Type: application/sdp
Content-Length: 142
```

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:1/1
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,50/50,5000,0
a=measurement:latency 30/30
a=measurement:jitter 6/4
a=measurement:bandwidth 0/4000
a=measurement:packetloss 0/3
=====
```

At this point there are two possibilities:

- o The bandwidth and packet loss constraints are not reached in one or both directions.
- o The bandwidth and packet loss constraints are reached in both directions.

3.1.2.2.1. Constraints not reached

If the measurements do not reach the quality constraints, the server answers with a 412 message (a precondition setting required by the client or server has not been met). Otherwise it returns 200 OK.

In the 412 message, the server may include a different value for the "qos-level" SDP session-level attribute, and the measurements of bandwidth and packet loss in both directions. All these information MUST be protected using the signature header.

Server Answer:

=====

Q4S/1.0 412 downlink_bandwidth

Date: Mon, 10 Jun 2010 10:00:01 GMT

Content-Type: application/sdp

Expires: 3000

Cause:downlink_bandwidth

Signature: 6eclba40e2adf2d783de530ae254acd4f3477ac4

Content-Length: 131

v=0

o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33

s=Q4S

i=Q4S parameters

t=0 0

a=qos-level:1/2

a=latency:40/35

a=jitter:10/10

a=bandwidth:20/6000

a=packetloss:5/5

a=flow:data downlink TCP/10000-20000

a=flow:control downlink UDP/55000

a=flow:control downlink TCP/55001

a=flow:data uplink TCP/56000

a=flow:control uplink UDP/56000

a=flow:control uplink TCP/56001

a=measurement:procedure default,50/50,50/50,5000,0

a=measurement:latency 30/30

a=measurement:jitter 6/4

a=measurement:bandwidth 200/4000

a=measurement:packetloss 2/3

=====

After a 412 message the client MUST send a QOS-ALERT request to acknowledge the SLA violation (using TCP control port). Notice that the server signature header is present in the client request, in order to allow integrity validation.

Client Request:

```
=====
QOS-ALERT q4s://www.example.com Q4S/1.0
Host: www.example.com
User-Agent: q4s-ua-experimental-1.0
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Type: application/sdp
Content-Length: 142
```

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:1/2
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,50/50,5000,0
a=measurement:latency 30/30
a=measurement:jitter 6/4
a=measurement:bandwidth 200/4000
a=measurement:packetloss 2/3
=====
```

Upon receiving the QOS-ALERT request from the client, the server will issue another QOS-ALERT request towards the client.

Server Answer:

```
=====
QOS-ALERT q4s://www.example.com Q4S/1.0
Date: Mon, 10 Jun 2010 10:00:01 GMT
Content-Type: application/sdp
Expires: 3000
Cause: latency
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Length: 131
```

```
v=0
o=q4s-UA 53655765 2353687637 IN IP4 192.0.2.33
s=Q4S
i=Q4S parameters
t=0 0
a=qos-level:1/2
a=latency:40/35
a=jitter:10/10
a=bandwidth:20/6000
a=packetloss:5/5
a=flow:data downlink TCP/10000-20000
a=flow:control downlink UDP/55000
a=flow:control downlink TCP/55001
a=flow:data uplink TCP/56000
a=flow:control uplink UDP/56000
a=flow:control uplink TCP/56001
a=measurement:procedure default,50/50,50/50,5000,0
a=measurement:latency 30/30
a=measurement:jitter 6/4
a=measurement:bandwidth 200/4000
a=measurement:packetloss 2/3
=====
```

After the client receives this request, both client and server wait for a while indicated in the server "Guard-Time" header, for example to allow different actions to be carried out by the server. (5 seconds should be enough, but this depends on each case) and begins again the measurement process of this stage (bandwidth and packet loss), starting from the beginning (sending the READY request). The maximum qos-level is 9/9 and if this value is reached without reaching the constraints, the quality session is aborted using the CANCEL method, which is detailed further in this document.

3.1.2.2.2. Constraints not reached with Policy server involved

If during the handshake phase the optional header Q4S-policy-server is included in the server response, the QOS-ALERT message MUST be sent to the policy server. The involved messages and operations are described in 2.1.2.1.2

3.1.2.2.3. Constraints reached

When measurements match the constraints, the server answer should be 200 OK, and MUST include the URI for triggering the application using an optional "Trigger-URI" header.

Server Answer:

=====

Q4S/1.0 200 OK

Date: Mon, 10 Jun 2010 10:00:01 GMT

Trigger-URI: http://www.example.com/app_start

Expires: 3000

Content-Type: application/sdp

Signature: 6eclba40e2adf2d783de530ae254acd4f3477ac4

Content-Length: 131

(SDP not shown)

=====

It means that client and server are ready to start the application.

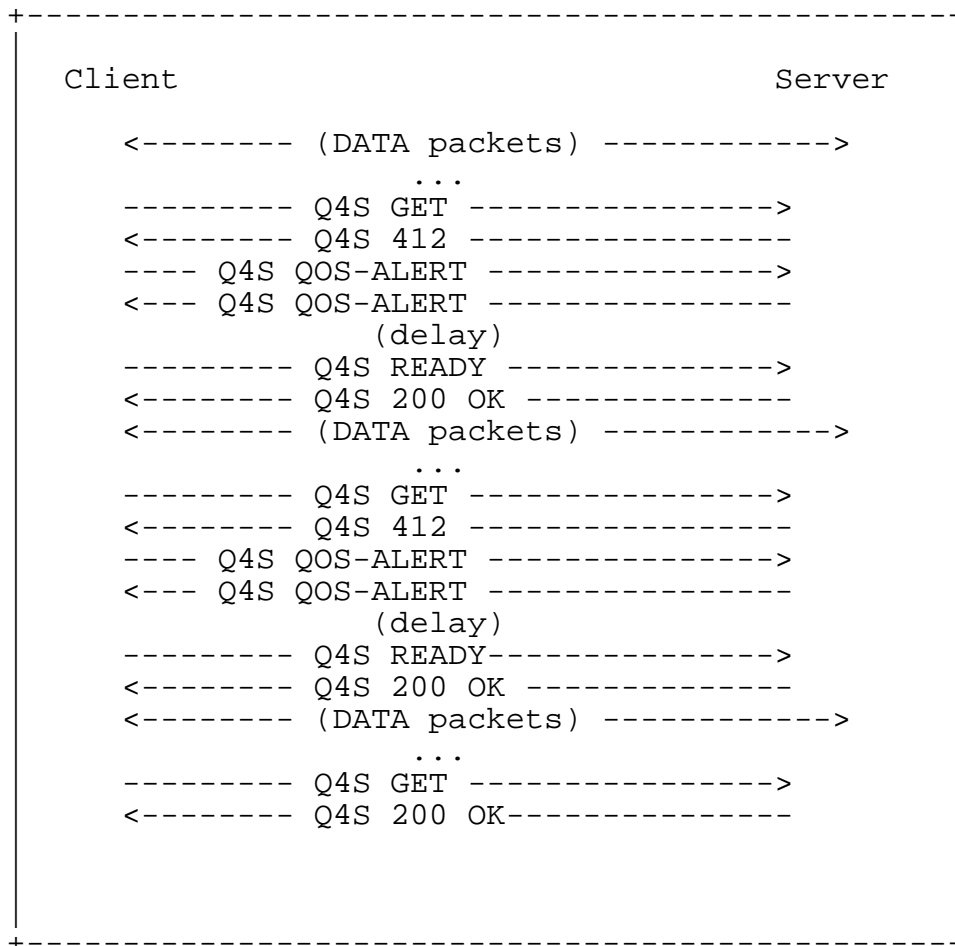


Figure 11 Bandwidth & packet loss measurement with success.

3.1.2.3. QoS Level out of range

If the qos-level has reached the maximum value for downlink or uplink without matching the constraints, then a CANCEL request MUST be sent in order to release the session. This request MUST be sent by the client using the control TCP port and does not have to be answered. In reaction to the receipt of the CANCEL request, the server MUST send a CANCEL request too. If the CANCEL request is not send to or received, the expiration time cancels the session at server side.

Client Request:

```
=====
CANCEL q4s://www.example.com Q4S/1.0
Host: www.example.com
User-Agent: q4s-ua-experimental-1.0
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Type: application/sdp
Content-Length: 142
```

(SDP not shown)

Server Request in reaction to Client Request:

```
=====
CANCEL q4s://www.example.com Q4S/1.0
Date: Mon, 10 Jun 2010 10:00:01 GMT
Expires: 0
Content-Type: application/sdp
Signature: 6ec1ba40e2adf2d783de530ae254acd4f3477ac4
Content-Length: 131
```

(SDP not shown)

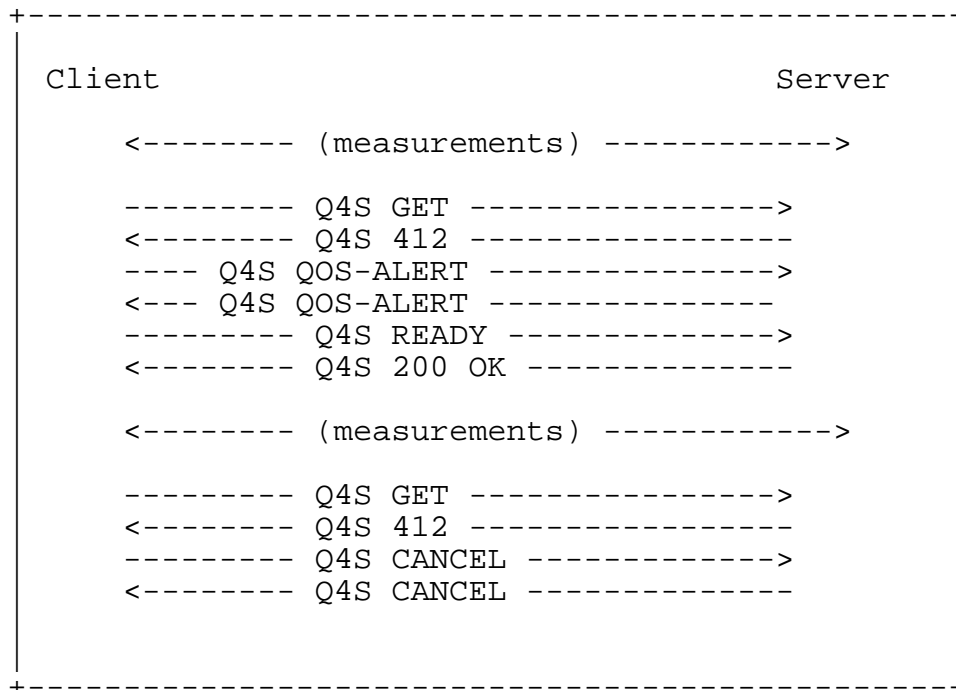


Figure 12 Failed negotiation phase.

3.1.2.4. QoS Level increments without changes in network behaviour

If the qos-level has not reached the maximum value (9) but after 3 QOS-ALERT messages (with increments in qos-level) the network remains with the same quality values, the client and the server MUST assume that the network can not reach the desired quality and abort the session in order to save resources (time and traffic). To do that, the client MUST send a CANCEL request and the server MUST react to it sending a CANCEL request too.

If the client does not send a CANCEL request but a request using a different method, the server MUST react to it sending a CANCEL request.

3.1.2.5. Trigger an application in combination with HTTP

When the negotiation phase is successful, an optional simple mechanism, based on HTTP, is defined to trigger the application.

The application may be triggered using an URI, by means of an HTTP request, just after negotiation success. The URI MUST be specified in the Q4S header "Trigger-URI". Other mechanisms, such as including

a "Location" header in the Q4S message, to force redirection is not recommended because these mechanisms are achieved without parsing the body of the message.

Example of use

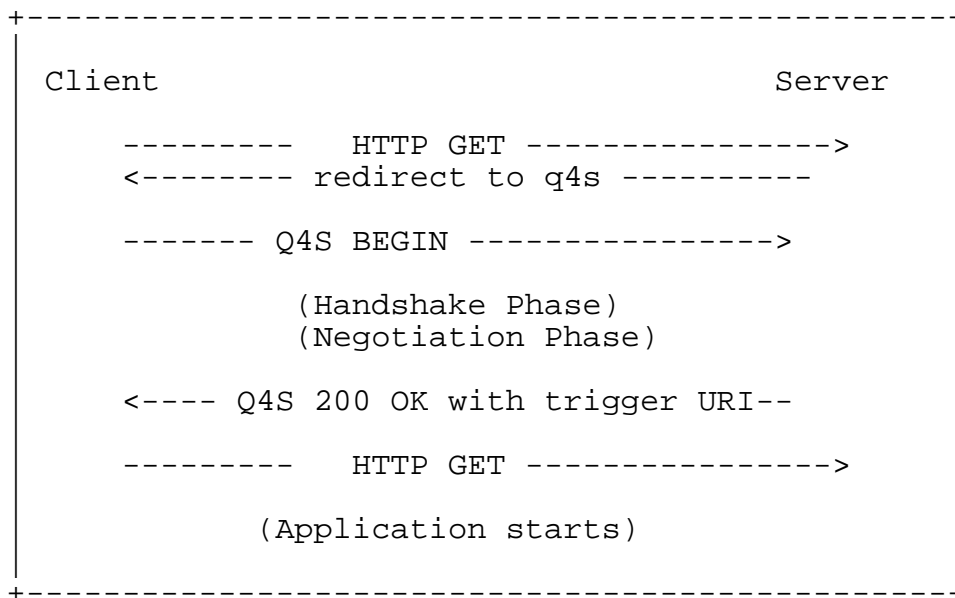


Figure 13 Trigger the application using HTTP URI

In the example, an integration of HTTP and Q4S is shown. First, the client contacts the server using HTTP, a redirection to a Q4S URI is achieved and the User Agent starts the Q4S handshake phase. After negotiation phase succeeds, the client trigger the application using the URI indicated in the Q4S 200 OK message.

3.1.3. Continuity phase

During the negotiation phase, latency, jitter, bandwidth and packet loss can be measured, but during continuity phase bandwidth will not be measured because bandwidth measurements may disturb application performance.

This phase is supposed to be executed at the same time as the real time application is being used.

In the default measurement procedure, two working modes are defined for this phase (normal and sliding window). The details of working modes are procedure dependant, and this draft only covers the default procedure.

3.1.3.1. Normal mode

The server can force the use of normal mode by setting the fourth parameter of "procedure" SDP attribute to 0. If this parameter is not set, the default value is assumed (zero), and normal mode will be used.

Example:

```
a=measurement:procedure default,50/50,50/50,5000,0
```

Considering that network conditions can change, the client may periodically check network conditions against negotiated constraints. The maximum interval expected between network testing is indicated in the Q4S Expires header.

However, the measurements can be carried out periodically in a smaller period of time than "Expires" header value. Intense interactive applications, like arcade videogames, the period to repeat the measurements may be very small (even zero), in order to measure continuously the quality and assure the best reaction time. To reach the best reaction time, the use of the sliding window mode is recommended.

To start the continuity phase, the client sends a Q4S READY method, using the TCP control port, exactly the same as Negotiation, indicating the new Stage header value for continuity phase (value 2).

Client Request:

```
=====
  READY q4s://www.example.com Q4S/1.0
  User-Agent: q4s-ua-experimental-1.0
  Stage: 2
  Session-Id: 53655765
  Content-Length: 0
=====
```

Server Response:

```
=====
  Q4S/1.0 200 OK
  Session-Id: 53655765
  Stage: 2
  Content-Length: 0
=====
```

After these messages are exchanged, latency, jitter and packet loss measurement are started, taking care of bandwidth usage. If the default measurement method is being used, it is recommended to use a larger interval for PING messages than the one used in the negotiation phase, but the same number of samples will be taken to check quality. The goal of incrementing the interval of PING messages is to minimize the load of the server which would be running lots of connections in parallel.

The process is the same as described in the negotiation phase. The difference is the time between samples, because the bandwidth usage MUST be protected. The interval used for this phase is indicated in the second parameter of the attribute line for the procedure. In this example, the interval is 75 milliseconds.

```
a=measurement:procedure default,50/50,75/75,5000,0
```

A value larger than the one used in the negotiation phase is recommended.

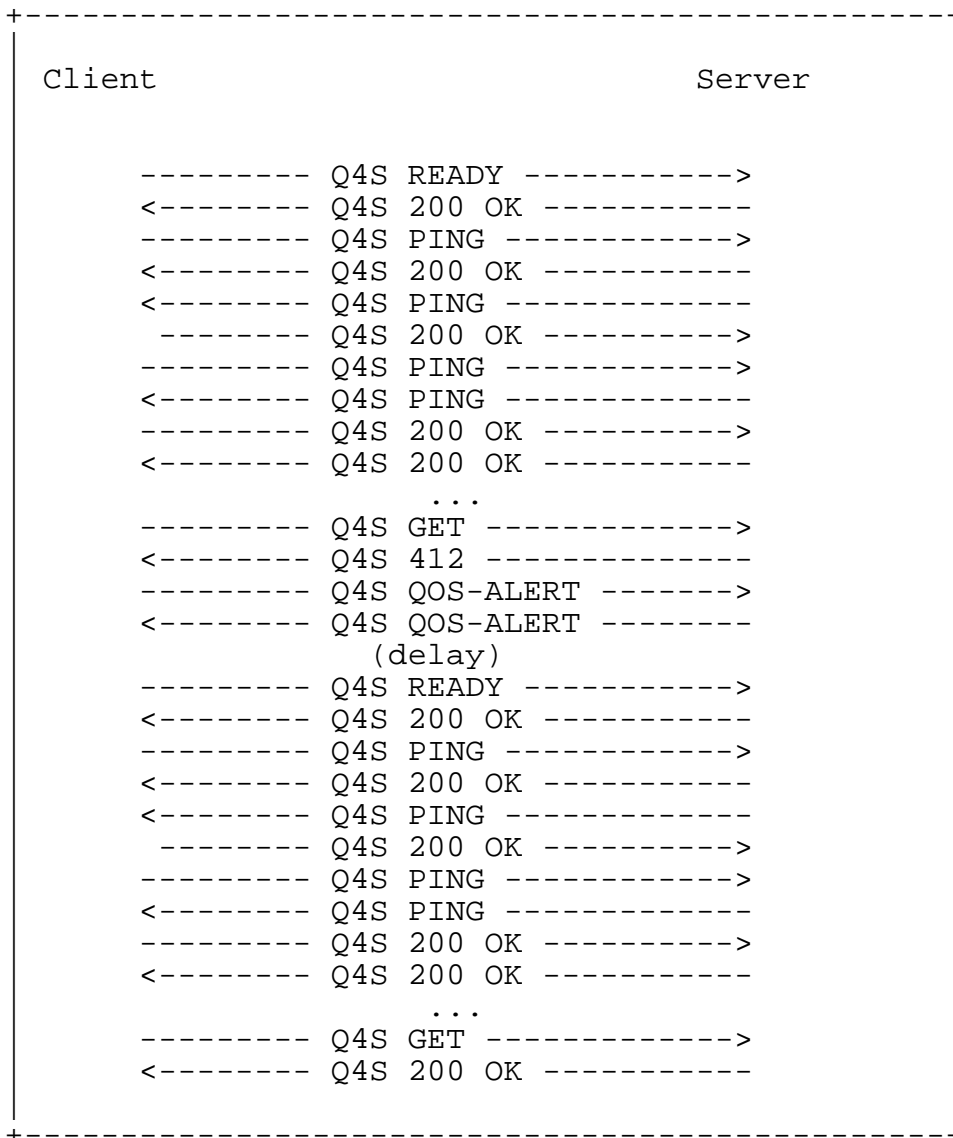


Figure 14 Continuity.

3.1.3.2. Sliding window mode

In order to improve the reaction time when network conditions degrade quickly, the server can force the use of the sliding window mode by setting the fourth parameter of the "procedure" SDP attribute to 1.

Example:

a=measurement:procedure default,50/50,50/50,5000,1

The sliding window mode applies a sliding window of 100 samples instead cycles of 100 samples.

In the sliding window mode, PING requests are sent continuously (in both directions) and when the Sequence-Number header reaches the value of 100, the client MUST NOT send a GET message for instructions, but continues sending PING messages with the Sequence-Number header starting again at zero. When the server PING Sequence-Number header reaches 100, it does the same, starting again from zero.

On the client side, the measured values of downlink jitter, downlink packet loss and latency are calculated using the last samples, discarding older ones, in a sliding window schema.

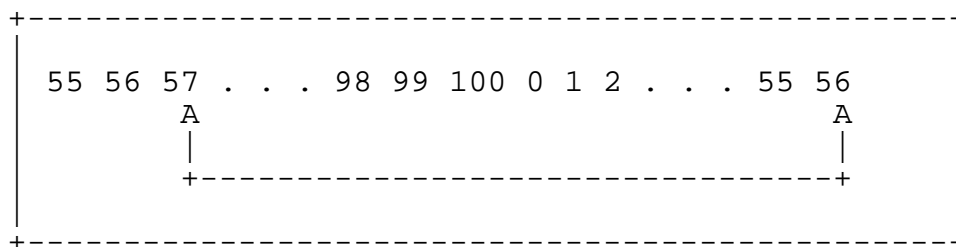


Figure 15 Sliding samples window

Only when the client detects that the measured values (downlink jitter, downlink packet loss and latency) are not reaching the constraints, a GET request is sent to the server.

When the server receives the GET request, it stops sending PING requests and answers the GET request just received. If the response code is 412, then a QOS-ALERT will be requested by the client, exactly in the same way as described in normal mode.

On the other hand, if the server detects that the measured values (uplink jitter, uplink packet loss and latency) are not reaching the constraints, it MUST choose between the following alternatives:

- o The server stops sending PING request to the client. In this case the client MUST notice this lack of PING requests using a timeout at reception. If so, the client reacts stopping the sending of PING requests to the server and sends a GET request for instructions, exactly in the same way as described in normal mode.

- o It continues sending PING requests but all of them with Sequence-Number set to -1 till a client GET request is received. Then the server stops sending PING messages and answers the GET request with the corresponding 412 error, exactly in the same way as described in normal mode. The client reacts sending this GET request when it receives a PING request with Sequence-Number header set to -1. This behavior allows the shortest reaction time under degradation of network conditions.

Both alternatives MUST be implemented by the Q4S client.

3.2. Dynamic constraints and flows

Depending on the nature of the application, the constraints to be reached may evolve, changing some or all constraint values in any direction.

The client MUST be able to deal with this possibility. When the server sends an SDP document attached to a reply (200 OK, or 412, etc), the client MUST assume all the new received values, overriding any previous value in use.

The dynamic changes on the constraints can be as a result of two possibilities:

- o The application communicates to the Q4S a change in the constraints. In this case the application requirements can evolve and the Q4S server will be aware of them.
- o The application uses TCP flows. In that case, in order to guarantee a constant throughput, the nature of TCP behavior forces the use of a composite constraint function which depends on RTT, packet loss and window control mechanism implemented in each TCP stack.

TCP throughput can be less than actual bandwidth if the Bandwidth-Delay Product (BDP) is large or if the network suffers from a high packet loss rate. In both cases, TCP congestion control algorithms may result in a suboptimal performance.

Different TCP congestion control implementations like Reno [14], High Speed TCP (RFC 3649 [15]), CUBIC [16], Compound TCP (CTCP [17]), etc. reach different throughputs under the same network conditions of RTT and packet loss. In all cases, depending on the RTT measured value, the Q4S server could change dynamically the packetloss constraints (defined in SDP) in order to make possible to

reach a required throughput or viceversa (use packetloss measurement to change dynamically latency constraints).

A general guideline to calculate the packetloss constraint and RTT constraint consists in approximating the throughput using a simplified formula which should take into account the TCP stack implementation of the receiver, in addition to RTT and packet loss:

$$Th = \text{Function}(RTT, \text{packet loss}, \dots)$$

Then, depending on RTT measured values, set dynamically the packetloss constraint.

It is possible to easily calculate a worst-case boundary for the Reno algorithm which should ensure for all algorithms that the target throughput is actually achieved. Except that, high-speed algorithms will then have even a larger throughput, if more bandwidth is available.

For the Reno algorithm, the Mathis' formula may be used [15] for the upper bound on the throughput:

$$Th \leq (MSS/RTT) * (1 / \sqrt{p})$$

In absence of packet loss, a practical limit for the TCP throughput is the receiver_window_size divided by the round-trip time. However, if the TCP implementation uses a window scale option, this limit can reach the available bandwidth value.

3.3. Qos-level downgrade operation

During the continuity phase it might be desirable to downgrade the current qos-level SDP parameter.

The strategy to carry out downgrades must include the possibility to exclude specific data flows from SDP dynamically. A Q4S client MUST allow this kind of SDP modifications by server.

Periodically (every several minutes, depending on the implementation) the server could force a QOS-ALERT, in which the level is downgraded for control flows, excluding application data flows from the embedded SDP of that request. To set the new SDP, the server MUST include the modified SDP in the 412 error message.

This mechanism allows to measure at lower levels of quality while application flows continue using a higher qos level value.

- o If the measurements in the lower level meet the constraints, then a new QOS-ALERT to this lower qos-level can be forced by the server, in which the SDP includes the application data flows in addition to control flows.
- o If the measurements in the lower level do not meet the constraints, then a new QOS-ALERT to the previous qos-level MUST be forced by the server, in which the SDP includes only the control flows.

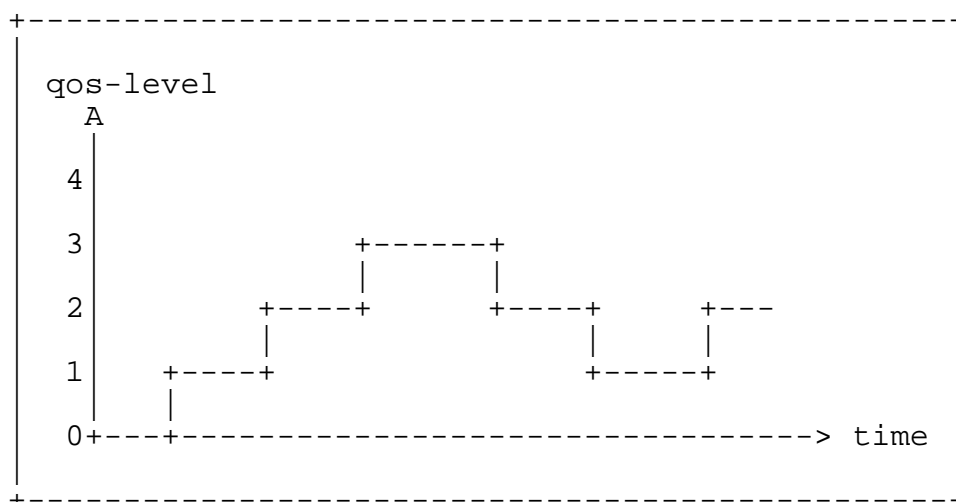


Figure 16 Possible evolution of qos-level

This mechanism avoids the risk of disturbing the application, while the measurements are being run in lower levels. However, this optimization of resources is optional, and MUST be used carefully.

The chosen period to measure a lower qos level is implementation dependant. Therefore it is not included as a measurement procedure parameter. It is recommended to use a large value, such as 20 minutes.

3.4. Sanity check of Quality sessions

A session may finish by several reasons (client shutdown, client CANCEL request, constraints not reached, etc), and any session finished MUST release the assigned resources.

In order to release the assigned server resources for the session, the "Expires" header indicates the maximum interval of time that a client can wait to repeat the continuity phase (in normal mode).

4. Q4S messages

Q4S is a text-based protocol and uses the UTF-8 charset (RFC 3629 [11]). A Q4S message is either a request or a response.

Both Request and Response messages use the basic format of Internet Message Format (RFC 5322 [12]). Both types of messages consist of a start-line, one or more header fields, an empty line indicating the end of the header fields, and an optional message-body.

```
generic-message = start-line
                  *message-header
                  CRLF
                  [ message-body ]
start-line      = Request-Line / Status-Line
```

The start-line, each message-header line, and the empty line MUST be terminated by a carriage-return line-feed sequence (CRLF). Note that the empty line MUST be present even if the message-body is not.

Much of Q4S's messages and header field syntax are identical to HTTP/1.1. However, Q4S is not an extension of HTTP.

4.1. Requests

Q4S requests are distinguished by having a Request-Line for a start-line. A Request-Line contains a method name, a Request-URI, and the protocol version separated by a single space (SP) character.

The Request-Line ends with CRLF. No CR or LF are allowed except in the end-of-line CRLF sequence. No linear whitespace (LWS) is allowed in any of the elements.

```
Request-Line = Method SP Request-URI SP Q4S-Version CRLF
```

Method: This specification defines five methods: GET for getting information and sending quality reports, PING and DATA for quality measurements purpose, CANCEL for terminating

sessions, and QOS-ALERT for querying ISPs for quality upgrades.

Request-URI: The Request-URI is a Q4S URI (RFC 2396) as described in 2.2.1. It normally indicates the user or service to which this request is being addressed to, but in the Q4S context, there are some methods whose URI only reflects the service on the server side, but nothing more. This is the case of the QOS-ALERT method, because the real address of a QoS upgrade request is the network, and therefore in this case the URI only reflects the server address. In addition the CANCEL method has the same treatment, and in the ECHO and DATA methods invoked by the server to the client the meaning of the URI is only the URI of the service, but not the destination of the request. The Request-URI MUST NOT contain unescaped spaces or control characters and MUST NOT be enclosed in "<>".

Q4S-Version: Both request and response messages include the version of Q4S in use. To be compliant with this specification, applications sending Q4S messages MUST include a Q4S-Version of "Q4S/1.0". The Q4S-Version string is case-insensitive, but implementations MUST send upper-case. Unlike HTTP/1.1, Q4S treats the version number as a literal string. In practice, this should make no difference.

4.2. Responses

Q4S responses are distinguished from requests by having a Status-Line as their start-line. A Status-Line consists of the protocol version followed by a numeric Status-Code and its associated textual phrase, with each element separated by a single SP character. No CR or LF is allowed except in the final CRLF sequence.

Status-Line = Q4S-Version SP Status-Code SP Reason-Phrase CRLF

The Status-Code is a 3-digit integer result code that indicates the outcome of an attempt to understand and satisfy a request. The Reason-Phrase is intended to give a short textual description of the Status-Code. The Status-Code is intended for use by automata, whereas the Reason-Phrase is intended for the human user. A client is not required to examine or display the Reason-Phrase.

While this specification suggests specific wording for the reason phrase, implementations MAY choose other text, for example, in the language indicated in the Accept-Language header field of the request.

The first digit of the Status-Code defines the class of response. The last two digits do not have any categorization role. For this reason, any response with a status code between 100 and 199 is referred to as a "1xx response", any response with a status code between 200 and 299 as a "2xx response", and so on. Q4S/1.0 allows following values for the first digit:

1xx: Provisional -- request received, continuing to process the request;

2xx: Success -- the action was successfully received, understood, and accepted;

3xx: Redirection -- further action needs to be taken in order to complete the request;

4xx: Client Error -- the request contains bad syntax or cannot be fulfilled at this server;

5xx: Server Error -- the server failed to fulfill an apparently valid request;

6xx: Global Failure -- the request cannot be fulfilled at any server.

The status codes are the same described in HTTP (RFC 2616 [1]). In the same way as HTTP, Q4S applications are not required to understand the meaning of all registered status codes, though such understanding is obviously desirable. However, applications MUST understand the class of any status code, as indicated by the first digit, and treat any unrecognized response as being equivalent to the x00 status code of that class.

The Q4S-ALERT and CANCEL requests do not have to be responded.

4.3. Header Fields

Q4S header fields are identical to HTTP header fields in both syntax and semantics.

Some header fields only make sense in requests or responses. These are called request header fields and response header fields, respectively. If a header field appears in a message not matching its category (such as a request header field in a response), it MUST be ignored.

4.3.1. Specific Q4S Request Header Fields

In addition to HTTP header fields, these are the specific Q4S request header fields

- o Session-Id: the value for this header is the same session id used in SDP and is assigned by the server. The messages without SDP MUST include this header. If a message has and SDP body, this header is optional. The method of <session id> allocation is up to the creating tool, but it is suggested that a UTC timestamp be used to ensure uniqueness.
- o Sequence-Number: sequential integer number assigned to PING and DATA messages.
- o Timestamp: this optional header contains the system time (with the best possible accuracy). Indicates the time in which the request was sent.
- o Signature: this header contains a digital signature that can be used by the network to validate the SDP. The signature is always generated by the server. It is optional.
- o Q4S-Resource-Client: this optional header contains the relative URI in charge of this session at the client side. In The case of being included, it MUST appear in the GET request of handshake phase. This URI MUST be invoked by the server in all later requests. It is optional, but it should be present, it becomes mandatory for the counterpart. This URI MUST be relative because user agents can not have associated domain, in addition to ignore their public IP address.

4.3.2. Specific Q4S Response Header Fields

- o Expires: the purpose is to provide a sanity check and allow the server to close inactive sessions. If the client does not send a new request before the expiration time, the server can close the session. The value MUST be an integer and the measurement unit are milliseconds.
- o Guard-time: A time interval in milliseconds left vacant (i.e., during which no data is sent) during the quality session. The guard time provides a safety margin before re-starting each measurement process when a QOS-ALERT has been raised. This header is optional in all messages but mandatory in the QOS-ALERT sent by the server.
- o Sequence-Number: same meaning as Request Header Fields
- o Timestamp: UTC time in nanoseconds. Indicates the time in which the request was sent. If the server (or a client) receives a Timestamp header in a request, MUST include the same header with the same value in the response. The purpose of this header is simplify the RTT calculation.
- o Signature: same meaning as Request Header Fields
- o Q4S-Resource-Server: this optional header contains the URI in charge of this session (Session URI). In case of being included, it MUST appear in the response to the BEGIN request of the handshake phase. This URI MUST be invoked by the client in all later requests. It is optional, but if present, it becomes mandatory for the counterpart.
- o Q4S-Policy-Server: this optional header contains the URI towards which the client and MUST send the QOS-ALERT messages (Policy Server URI). In case this header is present, the Q4S-Resource-Server header is mandatory, and MUST be included in the QOS-ALERT messages sent by the client to the policy server. In addition, the QOS-ALERT sent to the policy server MUST contain the header Q4S-Resource-client
- o Cause: This header field is a comma-separated list which contains the cause(s) for which the connection constraints were not reached after measurement process. Current defined values are:
 - Downlink_latency

- Uplink_latency
- Downlink_jitter
- Uplink_jitter
- Downlink_bw
- Uplink_bw

4.4. Bodies

Requests, including new requests defined in extensions to this specification, MAY contain message bodies unless otherwise noted. The interpretation of the body depends on the request method.

For response messages, the request method and the response status code determine the type and interpretation of any message body. All responses MAY include a body.

The Internet media type of the message body MUST be given by the Content-Type header field.

4.4.1. Encoding

The body MUST NOT be compressed. This mechanism is valid for other protocols such as HTTP and SIP (RFC 3261 [13]), but a compression/coding scheme will limit certain logical implementations of the way the request is parsed, thus, making the protocol concept more implementation dependant. In addition, bandwidth calculation may not be valid if compression is used. Therefore, the HTTP request header "Accept-Encoding" can not be used in Q4S with different values than "identity" and if it is present in a request, the server MUST ignore it. In addition, the response header "Content-Encoding" is optional, but if present, the unique permitted value is "identity".

The body length in bytes is provided by the Content-Length header field. The "chunked" transfer encoding of HTTP/1.1 MUST NOT be used for Q4S (Note: The chunked encoding modifies the body of a message in order to transfer it as a series of chunks, each one with its own size indicator.)

5. General User Agent behavior.

5.1. Roles

In order to allow peer to peer applications, a Q4S User Agent (UA) MUST be able to assume both client and server role. The role assumed depends on who sends the first message.

In a communication between two UAs, the UA that sends the Q4S BEGIN request in the first place, for starting the handshake phase, shall assume the client role.

If both UASs send the BEGIN request at the same time, they will wait for a random time to restart again.

Otherwise, an UA may be configured to act only as server (e.g., content provider's side).

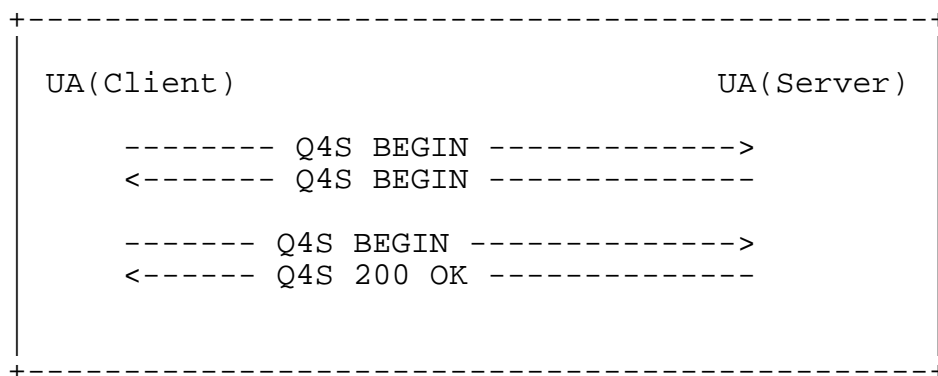


Figure 17 P2P roles.

5.2. Multiple Quality sessions in parallel

A quality session is intended to be used for a single application (or application instance). It means that for using the application, the client MUST establish only one quality session against the server. Indeed, the relation between Session-Id and application is 1 to 1.

If a user wants to raise several independent quality sessions simultaneously against different servers (or against the same server) it can execute multiple Q4S clients to establish separate quality sessions. However, this is not recommended because:

- o The establishment of a new quality session may affect other running applications over other quality sessions. Thus, minimum quality level may not be achieved depending on individual requirements of each application.
- o If the negotiation phase is executed separately before running any application, the quality requirements could not be assured when the applications are running in parallel.
- o Flow identification (Protocol, SourceIP, Source Port + Destination IP, Destination Port) must always be unique for each application/application instance, to ensure that each one of them is using their QoS constraints.

For running different applications in parallel it is highly recommended to execute the negotiation phase of all of them simultaneously, in order to assure the quality constraints of all applications in parallel. To do that, a single User Agent MUST be used, and this User Agent MUST be able to launch several quality session negotiations in parallel, synchronizing the beginning of each negotiation phase, and running again the negotiation phase of all applications in parallel until all of them succeed.

In order to repeat the execution of a negotiation phase that has been succeeded, both, client and server MUST allow using the READY method with a Stage header value already succeeded.

5.3. General client behavior

A Q4S Client has different behaviors. We will use letters X,Y,Z to designate each different behavior (follow the letter bullets in the figure below).

X) When it sends messages over TCP (methods GET, QOS-ALERT and CANCEL) it behaves strictly like a state machine that sends requests and waits for responses. Depending on the response type it enters in a new state.

When it sends UDP messages (methods PING and DATA), a Q4S client is not strictly a state machine that sends messages and waits for responses because:

Y) At latency, jitter and packet loss measurement, the PING requests are sent periodically, not after receiving the response to the previous request. In addition, the client MUST answer the

PING requests coming from the server, therefore assumes the role of a server.

Z) At bandwidth and packet loss measurement stage, the client does not expect to receive responses when sending DATA requests to the server. In addition, it MUST receive and process all server messages in order to achieve the downlink measurement.

The QOS-ALERT and CANCEL do not need to be answered. However, these methods may have a conventional answer if an error is produced.

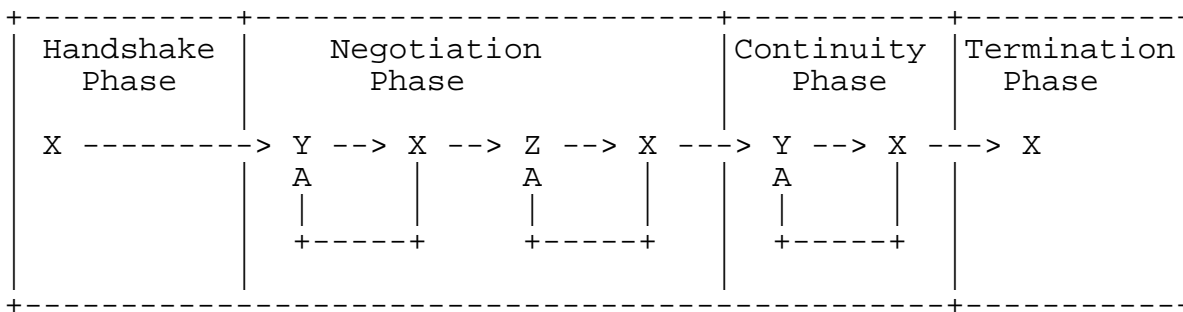


Figure 18 Phases & client behaviors.

5.3.1. Generating requests

A valid Q4S request formulated by a Client MUST, at a minimum, contain the following header fields:

If no SDP is included: This is the case of PING and DATA messages. The header Session-Id and Sequence-Number are mandatory.

If SDP is included: this is the case of GET, QOS-ALERT and CANCEL messages. Inside SDP is included Session-Id, therefore the inclusion of Session-Id header is optional.

5.4. General server behavior

If a server does not understand a header field in a request (that is, the header field is not defined in this specification or in any supported extension), the server MUST ignore that header field and continue processing the message.

The role of the server is changed at negotiation and continuity phases, in which server MUST send packets to measure jitter, latency and bandwidth. Therefore, the different behaviors of server are (follow the letter bullets in the figure below):

R) When the client sends messages over TCP (methods GET, QOS-ALERT and CANCEL) it behaves strictly like a state machine that receives messages and sends responses.

When the client begins to send UDP messages (methods PING and DATA), a QoS server is not strictly a state machine that receives messages and sends responses because:

S) At latency, jitter and packet loss measurement, the PING requests are sent periodically by the client but also by the server. In this case the server behaves as a server answering client requests but also behaves as a client, sending PING requests toward the client and receiving responses.

T) At bandwidth and packet loss measurement, the server sends DATA requests to the client. In addition, it MUST receive and process client messages in order to achieve the uplink measurement.

The QOS-ALERT and CANCEL do not need to be answered. However, these methods may have a conventional answer if an error is produced.

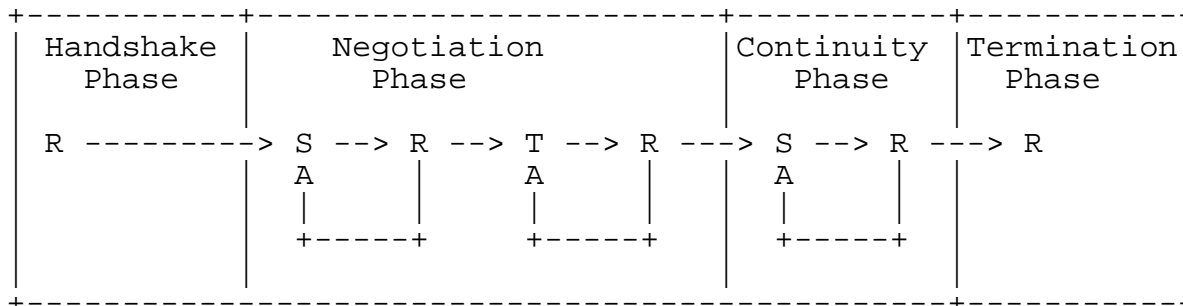


Figure 19 Phases & server behaviours.

6. Q4S method definitions

The Method token indicates the method to be performed on the resource identified by the Request-URI. The method is case-sensitive.

```
Method = "BEGIN" | "PING" | "DATA" | "GET" | "QOS-ALERT" |  
        "CANCEL" | "READY" | extension-method  
  
extension-method = token
```

The list of methods allowed by a resource can be specified in an "Allow" header field (RFC 2616 [1] section 14.7). The return code of the response always notifies the client when a method is currently allowed on a resource, since the set of allowed methods can change dynamically. Any server application SHOULD return the status code 405 (Method Not Allowed) if the method is known, but not allowed for the requested resource, and 501 (Not Implemented) if the method is unrecognized or not implemented by the server.

6.1. BEGIN

The BEGIN method means request information from a resource identified by a Q4S URI. The semantics of this method is the starting of a quality session.

This method is only used during the handshake phase to retrieve the SDP containing all quality parameters for the desired application to run.

When a BEGIN message is received by the server, any current quality session is cancelled and a new session should be created.

The response to a Q4S BEGIN request is not cacheable.

6.2. GET

The GET method means retrieve information from a resource identified by a Q4S URI.

In the negotiation and continuity phases, this method is used to check if the server considers the quality good enough to execute the desired application. If the measured quality is not enough, the server will return a 412 error.

The response to a Q4S GET request is not cacheable.

6.3. READY

The READY method is used to synchronize the starting time for sending of PING and DATA messages over UDP between clients and servers.

In addition, the Stage header included in this method is mandatory and allows clients to repeat a test, which is needed in scenarios with multiple quality sessions between one client and different servers.

This message is only used in negotiation and continuity phases, and only just before making a measurement. Otherwise (out of this context), the server MUST ignore this method.

6.4. PING

This message is used during the negotiation and continuity phases to measure the RTT and jitter of a session. The message MUST be sent only over UDP control port. If a server receives this message in another port it MUST ignore it.

The fundamental difference between the PING and DATA requests is reflected in the different measurements achieved with them. PING is a short message, and MUST be answered in order to measure RTT, whereas DATA is a long message (1 Kbyte) and MUST NOT be answered.

PING is a request method that can be originated by client but also by server. Client MUST answer the server PINGs, assuming a "server role" for these messages during measurement process.

6.5. DATA

This message is used only during the continuity phase to measure the bandwidth and packet loss of a session. The message MUST be sent only over UDP control port. If a server receives this message in other port it MUST ignore it.

The fundamental difference between the PING and DATA requests is reflected in the different measurements achieved with them. PING is a short message, and MUST be answered in order to measure RTT, whereas DATA is a long message (1 Kbyte) and MUST NOT be answered.

DATA is a request method that can be originated by the client but also by server. Both (client and server) MUST NOT answer DATA messages.

6.6. QOS-ALERT

This is the request message that Q4S generates when the measurements indicate that quality SLA is being violated. It is used during the negotiation and continuity phases.

This informative message indicates that the user experience is being degraded and includes the details of the problem (bandwidth, jitter, packet loss measurements and the SLA). The QOS-ALERT message does not contain any detail on the actions to be taken, which depends on the agreements between all involved parties.

A QOS-ALERT request does not have to be answered unless there is an error condition. However, after receiving a QOS-ALERT request, the server sends a QOS-ALERT request to the client.

This method can be initiated by the client only after a 412 error coming from server, and with enough information to build the QOS-ALERT message.

If the "Q4S-Policy-Server" header was included in the server response of the handshake phase, the QOS-ALERT message MUST be sent to the URI indicated in this header, otherwise the QOS-ALERT message MUST be sent to the server.

With policy server, the QOS-ALERT message sent by the client MUST contain the URIs of the server and the client to be contacted later by the policy server. Therefore the following headers MUST be included in the client request: "Q4S-Resource-Server" and "Q4S-Resource-Client".

The response to a Q4S QOS-ALERT request is not cacheable.

6.7. CANCEL

Like QOS-ALERT, this message is used for communication with the network resources. The semantics in this case is the release of the special resources assigned to the session.

In the same way as QOS-ALERT, CANCEL does not need to be answered. However, if the server receives a CANCEL message, it should send a new CANCEL request towards the client acknowledging the reception.

7. Response codes

QoS response codes are used for TCP and UDP. However, in UDP only the response code 200 is used.

7.1. 100 Trying

This response indicates that the request has been received by the next-hop server (the policy server) and that some unspecified action is being taken on behalf of this request (for example, a database is being consulted). This response, like all other provisional responses, stops retransmissions of a QoS-ALERT by the client.

7.2. 200 OK

The request has succeeded.

7.3. Redirection 3xx

3xx responses give information about the user's new location, or about alternative services that might be able to satisfy the request.

The requesting client SHOULD retry the request at the new address(es) given by the Location header field.

7.4. Request Failure 4xx

4xx responses are definite failure responses from a particular server. The client SHOULD NOT retry the same request without modification (for example, adding appropriate headers or SDP values). However, the same request to a different server might be successful.

7.4.1. 400 Bad Request

The request could not be understood due to malformed syntax. The Reason-Phrase SHOULD identify the syntax problem in more detail, for example, "Missing Sequence-Number header field".

7.4.2. 404 Not Found

The server has definitive information that the user does not exist at the domain specified in the Request-URI. This status is also returned if the domain in the Request-URI does not match any of the domains handled by the recipient of the request.

7.4.3. 405 Method Not Allowed

The method specified in the Request-Line is understood, but not allowed for the address identified by the Request-URI.

The response MUST include an Allow header field containing a list of valid methods for the indicated address.

7.4.4. 406 Not Acceptable

The resource identified by the request is only able of generating response entities that have content characteristics not acceptable according to the Accept header field sent in the request.

7.4.5. 408 Request Timeout

The server could not produce a response within a suitable amount of time, and the client MAY repeat the request without modifications at any later time

7.4.6. 412 A precondition has not been met

The server is indicating that the SLA is being violated.

7.4.7. 413 Request Entity Too Large

The server is refusing to process a request because the request entity-body is larger than the one that the server is willing or able to process. The server MAY close the connection to prevent the client from continuing the request.

7.4.8. 414 Request-URI Too Long

The server is refusing to process the request because the Request-URI is longer than the one that the server accepts.

7.4.9. 415 Unsupported Media Type

The server is refusing to process the request because the message body of the request is in a format not supported by the server for the requested method. The server MUST return a list of acceptable formats using the Accept, Accept-Encoding, or Accept-Language header field, depending on the specific problem with the content.

7.4.10. 416 Unsupported URI Scheme

The server cannot process the request because the scheme of the URI in the Request-URI is unknown to the server.

7.5. Server Failure 5xx

5xx responses are failure responses given when a server itself is having trouble.

7.5.1. 500 Server Internal Error

The server encountered an unexpected condition that prevented it from fulfilling the request. The client MAY display the specific error condition and MAY retry the request after several seconds.

7.5.2. 501 Not Implemented

The server does not support the functionality required to fulfill the request. This is the appropriate response when a Server does not recognize the request method and it is not capable of supporting it for any user.

Note that a 405 (Method Not Allowed) is sent when the server recognizes the request method, but that method is not allowed or supported.

7.5.3. 503 Service Unavailable

The server is temporarily unable to process the request due to a temporary overloading or maintenance of the server. The server MAY indicate when the client should retry the request in a Retry-After header field. If no Retry-After is given, the client MUST act as if it had received a 500 (Server Internal Error) response.

A client receiving a 503 (Service Unavailable) SHOULD attempt to forward the request to an alternate server. It SHOULD NOT forward any other requests to that server for the duration specified in the Retry-After header field, if present.

Servers MAY refuse the connection or drop the request instead of responding with 503 (Service Unavailable).

7.5.4. 504 Server Time-out

The server did not receive a timely response from an external server it accessed in attempting to process the request.

7.5.5. 505 Version Not Supported

The server does not support, or refuses to support, the Q4S protocol version that was used in the request. The server is indicating that it is unable or unwilling to complete the request using the same major version as the client, other than with this error message.

7.5.6. 513 Message Too Large

The server was unable to process the request since the message length exceeded its capabilities.

7.6. Global Failures 6xx

6xx responses indicate that a server has definitive information about a particular policy not satisfied for processing the request.

7.6.1. 600 session not exist

The Session-Id is not valid

7.6.2. 601 quality level not allowed

The QOS level requested is not allowed for the pair client/server

7.6.3. 603 Session not allowed

The session is not allowed due to some policy (number of sessions allowed for the server is exceeded, or the time band of the QOS-ALERT is not allowed for the pair client/server, etc)

7.6.4. 604 authorization not allowed

The policy server does not authorize the QOS-ALERT operation because any internal or external reason.

8. Implementation Recommendations

8.1. Default client constraints

To provide a default configuration, it would be good that the client had a configurable set of Quality headers in the browser settings menu. Otherwise these quality headers will not be present in the first message.

Different business models (out of scope of this proposal) may be achieved: depending on who pays for the quality session, the server can accept certain Client parameters sent in the first message, or force billing parameters on the server side.

8.2. Bandwidth measurements

In programming languages or Operating Systems with timers or limited clock resolution, it is recommended to use an approach based on several intervals to send messages of 1KB, in order to reach the required bandwidth consumption using a rate as close as possible to a constant rate.

For example, if the resolution is 1 millisecond, and the bandwidth to reach is 11Mbps, a good approach consists of sending:

- 1 message of 1KB every 1 millisecond +
- 1 message of 1KB every 3 milliseconds +
- 1 message of 1KB every 23 milliseconds

The number of intervals depends on required bandwidth and accuracy that the programmer wants to achieve.

8.3. Packet loss measurement resolution

Depending on application nature and network conditions, a packet loss resolution less than 1% may be needed. In such case, there is no limit to the number of samples used for this calculation. A tradeoff between time and resolution should be reached in each case. For example, in order to have a resolution of 1/10000, the last 10000 samples should be considered in the packetloss measured value.

The problem of this approach is the reliability of old samples. If the interval used between PING messages is 50ms, then to have a resolution of 1/1000 it takes 50 seconds and a resolution of 1/10000 takes 500 seconds (more than 8 minutes). The reliability of a packet loss calculation based on a sliding window of 8 minutes depends on how fast network conditions evolve.

8.4. Measurements and reactions

Q4S can be used as a mechanism for measure and trigger actions (i.e. lowering video bit-rate) in real-time in order to reach the application constraints, addressing measured possible network degradation.

The trigger is based on message QOS-ALERT, which is always forced by the server response 412 error. A server can avoid these QOS-REQUEST messages sending 200 OK when a GET message is received from server, independently whether the constraints are met or not.

8.5. Scenarios

Q4S could be used in two scenarios:

- o client to ACP (Application content provider)
- o client to client.

8.5.1. Client to ACP

In this scenario, the policy server is optional. If it exists, the QOS-ALERT messages MUST be sent to this policy server which acts as a proxy for this type of messages and validates them (plus any other actions out of scope of this document).

In order to avoid useless load on the server, the policy server could receive the BEGIN messages of handshake phase. For this purpose, the policy server MUST know the URI of the Q4S servers.

In this scenario a client could send the BEGIN to the policy server, with an additional parameter in the URI requested, which identifies the server, like:

```
Q4s://www.policy.com/listofservers?id=xtiwn28821ho4
```

Then the Policy Server validates the request and forward the BEGIN to the Q4S server, adding the Q4S-Resource-Server to the response for the client in the 200 OK response.

10. IANA Considerations

A specific port for Q4S TCP control flow mechanism could be assigned. It could simplify the network implementation. Other possibility is to use any other port (like 80, HTTP). In this case the network could use the protocol designator "Q4S" as the mark for distinguish and treat the packets.

Q4S uses SDP as a container for session information, in which quality attributes have been added as extended "session-level" attributes. These set of new attributes should be registered (in order to avoid the prefix "X-"). In this document, this set of attributes has been presented as registered attributes.

This is the list of attribute field names to register:

Attribute name: qos-level
Type of attribute: session level
Subject to the charset attribute: NO
Explanation of purpose: defines the current QoS profile in uplink and downlink for the communication between client and server. The exact meaning of each level is implementation dependant but in general, a higher qos-level value corresponds to a better quality network profile.
Appropriate attribute values: [0..9] "/" [0..9]

Attribute name: latency
Type of attribute: session level
Subject to the charset attribute: NO
Explanation of purpose: defines the latency constraints in milliseconds in uplink and downlink for the communication between client and server. Appropriate attribute values: [0..9999] "/" [0..9999]
If there is no constraint in some direction (uplink, downlink or both) the value can be empty in that direction

Attribute name: jitter
Type of attribute: session level
Subject to the charset attribute: NO
Explanation of purpose: defines the jitter constraints in milliseconds in uplink and downlink for the communication between client and server.
Appropriate attribute values: [0..9999] "/" [0..9999]

Attribute name: bandwidth
Type of attribute: session level
Subject to the charset attribute: NO
Explanation of purpose: define the bandwidth constraints in kbps in uplink and downlink for the communication between client and server.
Appropriate attribute values: [0..99999] "/" [0..99999]

Attribute name: packetloss
Type of attribute: session level
Subject to the charset attribute: NO
Explanation of purpose: define the packet loss tolerance constraints in 100% in uplink and downlink for the communication between client and server.
Appropriate attribute values: [0..99] "/" [0..99]

Attribute name: flow
Type of attribute: session level
Subject to the charset attribute: NO

Explanation of purpose: define a flow between a client and a server. The flow involves purpose (data or control), direction (uplink or downlink) protocol (UDP or TCP) and port or range or ports

Attribute values:

```
<"control"|"data"> <"uplink"|"downlink"> <"UDP"|"TCP">
<0..65535>[ "-" [0..65535]]
```

Attribute name: measurement

Type of attribute: session level

Subject to the charset attribute: NO

Explanation of purpose: define the procedure to measure the quality and the different values for each measurement

```
Attribute values: "procedure/" <procedure> |
                  "latency "[0..9999] "/" [0..9999] |
                  "jitter "[0..9999] "/" [0..9999] |
                  "bandwidth "[0..99999] "/" [0..99999] |
                  "packetloss "[0..99] "/" [0..99]
```

If the attribute value is "procedure", the rest of the line MUST contain the name of the procedure and optional parameters, separated by ",".

In the case of procedure "default", the valid values are:

```
a=measurement:procedure default,[0..999]"/" [0..999] ", " [0..999]
"/" [0..999] ", " [0..9999] ", " [0|1]
```

where:

- o The first parameter is the interval of time (in milliseconds) between PING messages in the negotiation phase. Forward (client to server) and reverse (server to client) values separated by "/".
- o The second parameter is the interval of time (in milliseconds) between PING messages in the continuity phase. Forward (client to server) and reverse (server to client) values separated by "/".
- o The third parameter is the time used to measure bandwidth during negotiation phase. In case of not present, a default value of 5000 ms will be assumed.
- o The fourth parameter indicates the mode for continuity phase (0 means "normal" and 1 means "sliding window"). In case of not be present, normal mode (default value of 0) will be assumed.

Other procedure names are allowed, but at least "default" procedure implementation is mandatory in client and servers.

11. Conclusions

Q4S defines four phases with different purposes, and inside these phases the negotiated measurement procedure is used. Different measurement procedures can be used (even RTCP itself) inside Q4S. Basically, Q4S only defines how to transport SLA information and measurement results as well as providing some mechanisms for alerting. Q4S does not ask for resources. Q4S only alerts if one (or some) of SLA quality parameters are being violated. Depends on server (Application content provider) to do something with this information and return it back to a SLA-compliant state.

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