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Media Gateway Control Protocol (MGCP) Mauricio Arango, Andrew Dugan, Isaac Elliott, Christian Huitema, Scott Pickett Version 0.1 draft

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Abstract

This document describes an application programming interface and a corresponding protocol (MGCP) for controlling Voice over IP (VoIP) Gateways from external call control elements. The MGCP assumes a call control architecture where the call control 'intelligence' is outside the gateways and handled by external call control elements.

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The document is structured in 6 main sections:

- * The introduction presents the basic assumptions and the relation to other protocols such as H.323, RTSP, SAP or SIP.
- * The interface section presents a conceptual overview of the MGCP, presenting the naming conventions, the usage of the session description protocol SDP, and the five procedures that compose MGCP: Notifications Request, Notification, Create Connection, Modify Connection, Delete Connection, AuditEndpoint, AuditConnection and RestartInProgress.
- * The protocol description section presents the MGCP encodings, which are based on simple text formats, and the transmission procedure over UDP.
- * The security section presents the security requirement of MGCP, and its usage of IP security services (IPSEC).
- * The event packages section provides an initial definition of packages and event names.
- * The description of the changes made in combining SGCP 1.1 and IPDC to create MGCP 0.1.

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1. Introduction

This document describes an abstract application programming interface and a corresponding protocol (MGCP) for controlling Telephony Gateways from external call control elements called media gateway controllers or call agents. A telephony gateway is a network element that provides conversion between the audio signals carried on telephone circuits and data packets carried over the Internet or over other packet networks. Example of gateways are:

- * Trunking gateways, that interface between the telephone network and a Voice over IP network. Such gateways typically manage a large number of digital circuits.
- * Voice over ATM gateways, which operate much the same way as voice over IP trunking gateways, except that they interface to an ATM network.
- * Residential gateways, that provide a traditional analog (RJ11) interface to a Voice over IP network. Examples of residential gateways include cable modem/cable set-top boxes, xDSL devices, broadband wireless devices
- * Access gateways, that provide a traditional analog (RJ11) or digital PBX interface to a Voice over IP network. Examples of access gateways include small-scale voice over IP gateways.
- * Business gateways, that provide a traditional digital PBX interface or an integrated "soft PBX" interface to a Voice over IP network.
- * Network Access Servers, that can attach a "modem" to a telephone circuit and provide data access to the Internet. We expect that, in the future, the same gateways will combine Voice over IP services and Network Access services.
- * Circuit switches, or packet switches, which can offer a control interface to an external call control element.

MGCP assumes a call control architecture where the call control "intelligence" is outside the gateways and handled by external call control elements. The MGCP assumes that these call control elements, or Call Agents, will synchronize with each other to send coherent commands to the gateways under their control. MGCP does not define a mechanism for synchronizing Call Agents. MGCP is, in essence, a master/slave protocol, where the gateways are expected to execute commands sent by the Call Agents. In consequence, this document specifies in great detail the expected behavior of the gateways, but only specify those parts of a call agent implementation, such as timer management, that are mandated

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for proper operation of the protocol.

MGCP assumes a connection model where the basic constructs are endpoints and connections. Endpoints are sources or sinks of data and could be physical or virtual. Examples of physical endpoints are:

- * An interface on a gateway that terminates a trunk connected to a PSTN switch (e.g., Class 5, Class 4, etc.). A gateway that terminates trunks is called a trunk gateway.
- * An interface on a gateway that terminates an analog POTS connection to a phone, key system, PBX, etc. A gateway that terminates residential POTS lines (to phones) is called a residential gateway.

An example of a virtual endpoint is an audio source in an audio-content server. Creation of physical endpoints requires hardware installation, while creation of virtual endpoints can be done by software.

Connections may be either point to point or multipoint. A point to point connection is an association between two endpoints with the purpose of transmitting data between these endpoints. Once this association is established for both endpoints, data transfer between these endpoints can take place. A multipoint connection is established by connecting the endpoint to a multipoint session.

Connections can be established over several types of bearer networks:

- * Transmission of audio packets using RTP and UDP over a TCP/IP network.
- * Transmission of audio packets using AAL2, or another adaptation layer, over an ATM network.
- * Transmission of packets over an internal connection, for example the TDM backplane or the interconnection bus of a gateway. This is used, in particular, for "hairpin" connections, connections that terminate in a gateway but are immediately rerouted over the telephone network.

For point-to-point connections the endpoints of a connection could be in separate gateways or in the same gateway.

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1.1. Relation with the H.323 standards

MGCP is designed as an internal protocol within a distributed system that appears to the outside as a single VoIP gateway. This system is composed of a Call Agent, that may or may not be distributed over several computer platforms, and of a set of gateways. In a typical configuration, this distributed gateway system will interface on one side with one or more telephony (i.e. circuit) switches, and on the other side with H.323 conformant systems, as indicated in the following table:

Functional Plane	Phone switch	Terminating Entity	H.323 conformant systems
Signaling Plane	Signaling exchanges through SS7/ISUP	Call agent	Signaling exchanges with the call agent through H.225/RAS and H.225/Q.931.
			Possible negotiation of logical channels and transmission parameters through H.245 with the call agent.
		Internal synchronization through MGCP	
Bearer Data Transport Plane	Connection through high speed trunk groups	Telephony gateways	Transmission of VOIP data using RTP directly between the H.323 station and the gateway.

In the MGCP model, the gateways focus on the audio signal translation function, while the Call Agent handles the signaling and call processing functions. As a consequence, the Call Agent implements the "signaling" layers of the H.323 standard, and presents itself as an "H.323 Gatekeeper" or as one or more "H.323 Endpoints" to the H.323 systems.

1.2. Relation with the IETF standards

While H.323 is the recognized standard for VoIP terminals, the IETF has also produced specifications for other types of multi-media

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applications. These other specifications include:

- * the Session Description Protocol (SDP), RFC 2327,
- * the Session Announcement Protocol (SAP),
- * the Session Initiation Protocol (SIP),
- * the Real Time Streaming Protocol (RTSP), RFC 2326.

The latter three specifications are in fact alternative signaling standards that allow for the transmission of a session description to an interested party. SAP is used by multicast session managers to distribute a multicast session description to a large group of recipients, SIP is used to invite an individual user to take part in a point-to-point or unicast session, RTSP is used to interface a server that provides real time data. In all three cases, the session description is described according to SDP; when audio is transmitted, it is transmitted through the Real-time Transport Protocol, RTP.

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The distributed gateway systems and MGCP will enable PSTN telephony users to access sessions set up using SAP, SIP or RTSP. The Call Agent provides for signaling conversion, according to the following table:

Functional Plane	Phone switch	Terminating Entity	IETF conforming systems
Signaling Plane	Signaling exchanges through SS7/ISUP	Call agent	Signaling exchanges with the call agent through SAP, SIP or RTSP.
			Negotiation of session description parameters through SDP (telephony gateway terminated but passed via the call agent to and from the IETF conforming system)
		Internal synchronization through MGCP	
Bearer Data Transport Plane	Connection through high speed trunk groups	Telephony gateways	Transmission of VoIP data using RTP, directly between the remote IP end system and the gateway.

The SDP standard has a pivotal status in this architecture. We will see in the following description that we also use it to carry session descriptions in MGCP.

1.3. Definitions

Trunk: A communication channel between two switching systems. E.g., a DS0 on a T1 or E1 line.

2. Media Gateway Control Interface

The interface functions provide for connection control and endpoint control. Both use the same system model and the same naming conventions.

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2.1. Model and naming conventions.

The MGCP assumes a connection model where the basic constructs are endpoints and connections. Connections are grouped in calls. One or more connections can belong to one call. Connections and calls are set up at the initiative of one or several Call Agents.

2.1.1. Names of endpoints

Endpoints names have two components:

- * the domain name of the gateway that is managing the endpoint ,
- * a local name within that gateway,

The syntax of the local name depends on the type of endpoint being named. However, the local name for each of these types is naturally hierarchical, beginning with a term which identifies the physical gateway containing the given endpoint and ending in a term which specifies the individual endpoint concerned. With this in mind, the following rules for construction and interpretation of the Entity Name field for these entity types MUST be supported:

- The individual terms of the naming path MUST be separated by a single slash ("/", ASCII 2F hex).
- 2) The individual terms are character strings composed of letters, digits or other printable characters, with the exception of characters used as delimitors ("/", "@"), characters used for wildcarding ("*", "\$") and white spaces.
- 3) Wild-carding is represented either by an asterisk ("*") or a dollar sign ("\$") for the terms of the naming path which are to be wildcarded. Thus, if the full naming path looks like

term1/term2/term3

then the Entity Name field looks like this depending on which terms are wild-carded:

/term2/term3 if term1 is wild-carded term1//term3 if term2 is wild-carded term1/term2/* if term3 is wild-carded term1/*/* if term2 and term3 are wild-carded, etc.

In each of these examples a dollar sign could have appeared instead

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of an asterisk.

4) A term represented by an asterisk is to be interpreted as: "use ALL values of this term known within the scope of the Media Gateway". A term represented by a dollar sign is to be interpreted as: "use ANY ONE value of this term known within the scope of the Media Gateway". The description of a specific command or AVP may add further criteria for selection within the general rules given here.

If the Media Gateway controls multiple physical gateways, the first term of the naming MUST identify the physical gateway containing the desired entity. If the Media Gateway controls only a single physical gateway, the first term of the naming string MAY identify that physical gateway, depending on local practice. A local name that is composed of only a wildcard character refers to either all (*) or any (\$) endpoints within the media gateway.

In the case of trunking gateways, endpoints are trunk circuits linking a gateway to a telephone switch. These circuits are typically grouped into a digital multiplex, that is connected to the gateway by a physical interface. Such circuits are named in three contexts:

- * In the ISUP protocol, trunks are grouped into trunk groups, identified by the SS7 point codes of the switches that the group connects. Circuits within a trunk group are identified by a circuit number (CIC in ISUP).
- * In the gateway configuration files, physical interfaces are typically identified by the name of the interface, an arbitrary text string. When the interface multiplexes several circuits, individual circuits are typically identified by a circuit number.
- * In MGCP, the endpoints are identified by an endpoint name.

The Call Agents use configuration databases to map ranges of circuit numbers within an ISUP trunk group to corresponding ranges of circuits in a multiplex connected to a gateway through a physical interface. The gateway will be identified, in MGCP, by a domain name. The local name will be structured to encode both the name of the physical interface, for example X35V3+A4, and the circuit number within the multiplex connected to the interface, for example 13. The circuit number will be separated from the name of the interface by a fraction bar, as in:

X35V3+A4/13

Other types of endpoints will use different conventions. For example, in gateways were physical interfaces by construction only control one

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circuit, the circuit number will be omitted. The exact syntax of such names should be specified in the corresponding server specification.

2.1.2. Names of calls

Calls are identified by unique identifiers, independent of the underlying platforms or agents. These identifiers are created by the Call Agent. They are treated in MGCP as unstructured octet strings.

Call identifiers are expected to be unique within the system. When a Call Agent builds several connections that pertain to the same call, either on the same gateway or in different gateways, these connections will all be linked to the same call through the globally unique identifier. This identifier can then be used by accounting or management procedures, which are outside the scope of MGCP.

2.1.3. Names of connections

Connection identifiers are created by the gateway when it is requested to create a connection. They identify the connection within the context of an endpoint. They are treated in MGCP as unstructured octet strings. The gateway should make sure that a proper waiting period, at least 3 minutes, elapses between the end of a connection that used this identifier and its use in a new connection for the same endpoint. (Gateways may decide to use identifiers that are unique within the context of the gateway.)

2.1.4. Names of Call Agents and other entities

The media gateway control protocol has been designed to allow the implementation of redundant Call Agents, for enhanced network reliability. This means that there is no fixed binding between entities and hardware platforms or network interfaces.

Reliability can be improved by the following precautions:

- * Entities such as endpoints or Call Agents are identified by their domain name, not their network addresses. Several addresses can be associated with a domain name. If a command or a response cannot be forwarded to one of the network addresses, implementations should retry the transmission using another address.
- * Entities may move to another platform. The association between a logical name (domain name) and the actual platform are kept in the domain name service. Call Agents and Gateways should keep track of the time-to-live of the record they read from the DNS. They should query the DNS to refresh the information if the time to live has expired.

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2.1.5. Digit maps

The Call Agent can ask the gateway to collect digits dialed by the user. This facility is intended to be used with residential gateways to collect the numbers that a user dials; it may also be used with trunking gateways and access gateways alike, to collect the access codes, credit card numbers and other numbers requested by call control services.

An alternative procedure is for the gateway to notify the Call Agent of the dialed digits, as soon as they are dialed. However, such a procedure generates a large number of interactions. It is preferable to accumulate the dialed numbers in a buffer, and to transmit them in a single message.

The problem with this accumulation approach, however, is that it is hard for the gateway to predict how many numbers it needs to accumulate before transmission. For example, using the phone on our desk, we can dial the following numbers:

0	Local operator
00	Long distance operator
xxxx	Local extension number
8xxxxxxx	Local number
#xxxxxxx	Shortcut to local number at
	other corporate sites
*xx	Star services
91xxxxxxxxxx	Long distance number
9011 + up to 15 digits	International number

The solution to this problem is to load the gateway with a digit map that correspond to the dial plan. This digit map is expressed using a syntax derived from the Unix system command, egrep. For example, the dial plan described above results in the following digit map:

(OT | 00T | [1-7]xxx | 8xxxxxxx | #xxxxxxx | *xx | 91xxxxxxxxx | 9011x.T)

The formal syntax of the digit map is described by the following BNF notation:

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```
Digit ::= "0" | "1" | "2" | "3" | "4" | "5" | "6" |
"7" | "8" | "9"
Timer ::= "T" -- matches the detection of a timer
Letter::= Digit | Timer | "#" | "*" | "A" | "B" | "C"
          "D"
Range ::= "x" -- matches any digit
         | "[" Letters "]" -- matches any of the specified letters
Letters ::= Subrange | Subrange Letters
Subrange ::= Letter -- matches the specified letter
  | Digit "-" Digit -- matches any digit between first and
                    -- last
Position ::= Letter
           Range
StringElement::= Position -- matches an occurrence of
                          -- the position
           | Position "." -- matches an arbitrary number of
                          -- occurrences
                          -- of the position, including 0
String::= StringElement | StringElement String
StringList::= String | String "|" StringList
DigitMap::= String | "(" StringList ")"
```

A DigitMap, according to this syntax, is defined either by a "string" or by a list of strings. Each string in the list is an alternative numbering scheme. A gateway that detects digits, letters or timers will:

- 1) Add the event parameter code as a token to the end of an internal state variable called the "current dial string"
- 2) Apply the current dial string to the digit map table, attempting a match to each regular expression in the Digit Map in lexical order
- 3) If the result is under-qualified (partially matches at least one entry in the digit map), do nothing further.

If the result matches, or is over-qualified (i.e. no further digits could possibly produce a match), send the current digit string to the Call Agent.

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Digit maps are provided to the gateway by the Call Agent, whenever the Call Agent instructs the gateway to listen for digits.

2.1.6. Names of events

The concept of events and signals is central to MGCP. A Call Agent may ask to be notified about certain events occurring in an endpoint, e.g. off-hook events, and a call agent may request certain signals to be applied to an endpoint, e.g. dial-tone.

Events and signals are grouped in packages within which they share the same namespace which we will refer to as event names in the following. Packages are groupings of the events and signals supported by a particular type of endpoint. For instance, one package may support a certain group of events and signals for analog access lines, and another package may support another group of events and signals for video lines. One or more packages may exist for a given endpoint-type.

Event names are case insensitive and are composed of two logical parts, a package name and an event name. Both names are strings of letters, hyphens and digits, with the restriction that hyphens shall never be the first or last characters in a name. Package or event names are not case sensitive - values such as "hu", "Hu", "HU" or "hU" should be considered equal.

Examples of package names are "D" (DTMF), "M" (MF), "T" (Trunk) or "L" (Line). Examples of event names can be "hu" (off hook or "hang-up" transition), "hf" (flash hook) or "0" (the digit zero).

In textual representations, the package name, when present, is separated from the event name by a slash ("/"). The package name is in fact optional. Each endpoint-type has a default package associated with it, and if the package name is excluded from the event name, the default package name for that endpoint-type is assumed. For example, for an analog access line, the following two event names are equal:

1/dl dial-tone in the line package for an analog access line.

dl dial-tone in the line package (default) for an analog access line.

This document defines a basic set of package names and event names. Additional package names and event names can be registered with the IANA. A package definition shall define the name of the package, and the definition of each event belonging to the package. The event definition shall include the precise name of the event (i.e., the code used in MGCP), a plain text definition of the event, and, went appropriate, the precise definition of the corresponding signals, for example the exact frequencies of audio signal such as dial tones or DTMF tones.

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In addition, implementers can gain experience by using experimental packages. The names of experimental packages must start with the two characters "x-"; the IANA shall not register package names that start with these characters.

Digits, or letters, are supported in many packages, notably "DTMF", "MF" and "pulse". Digits and letters are defined by the rules "Digit" and "Letter" in the definition of digit maps. These definition refers to the digits (0 to 9), to the asterisk ("*") and orthothrop or pound sign ("#"), and to the letters "A", "B", "C" and "D", as well as the timer indication "T". These letters can be combined in "digit string" that represent the keys that a user punched on a dial. In addition, the letter "X" can be used to represent all digits, and the sign "\$" can be used in wildcard notations. The need to easily express the digit strings has a consequence on the form of event names:

An event name that does not denote a digit should always contain at least one character that is neither a digit, nor one of the letters A, B, C, D, T or X. (Such names should not contain the special signs "*", "#", "/" or "\$".)

A Call Agent may often have to ask a gateway to detect a group of events. Two conventions can be used to denote such groups:

- * The wildcard convention can be used to detect any event belonging to a package, or a given event in many packages, or event any event in any package supported by the gateway.
- * The regular expression Range notation can be used to detect a range of digits.

The dollar sign ("\$") can be used as a wildcard instead of a package name or an event name:

A name such as "foo/\$" denotes all events in package "foo" A name such as "\$/bar" denotes the event bar in any package supported by the gateway The names "\$" or "\$/\$" denote all events supported by the gateway.

The call agent can ask a gateway to detect a set of digits or letters either by individually decribing those letters, or by using the "range" notation defined in the syntax of digit strings. For example, the call agent can:

Use the letter "x" to denote "any letter or digit." Use the notation "[0-9#]" to denote the digits 0 to 9 and the pound sign.

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Events and signals are described in packages. The package description must provide, for each events, the following informations:

- * The description of the event and its purpose,
- * The detailed characteristics of the event, such as for example frequencies and amplitude of audio signals, modulations and repetitions,
- * The typical and maximum duration of the event.

Signals are divided into different types depending on their behavior:

- * On/off (OO) Once applied, these signals last forever until they are turned off. This may happen either as the result of an event or a new SignalRequests (see later).
- * Time-out (TO) Once applied, these signals last until they are either turned off (by an event or SignalRequests) or a signal specific period of time has elapsed. Currently, a signal that times out does not generate any events.
- * Brief (BR) The duration of these signals is so short, that they stop on their own. If an event occurs the signal will not stop, however if a new SignalRequests is applied, the signal will stop.

TO signals are normally used to alert the endpoints' users, to signal them that they are expected to perform a specific action, such as hang down the phone (ringing). Transmission of these signals should typically be interrupted as soon as the first of the requested events has been produced.

Package descriptions should describe, for all signals, their type (OO, TO, BR). They should also describe the maximum duration of the TO signals.

2.2. Usage of SDP

The Call Agent uses the MGCP to provision the gateways with the description of connection parameters such as IP addresses, UDP port and RTP profiles. These descriptions will follow the conventions delineated in the Session Description Protocol which is now an IETF proposed standard, documented in RFC 2327.

SDP allows for description of multimedia conferences. This version

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limits SDP usage to the setting of audio circuits and data access circuits. The initial session descriptions contain the description of exactly one media, of type "audio" for audio connections, "nas" for data access.

2.3. Gateway Control Functions

This section describes the commands of the MGCP. The service consists of connection handling and endpoint handling commands. There are eight commands in the protocol:

- * The Call Agent can issue a NotificationRequest command to a gateway, instructing the gateway to watch for specific events such as hook actions or DTMF tones on a specified endpoint .
- * The gateway will then use the Notify command to inform the Call Agent when the requested events occur.
- * The Call Agent can use the CreateConnection command to create a connection that terminates in an "endpoint" inside the gateway.
- * The Call Agent can use the ModifyConnection command to change the parameters associated to a previously established connection.
- * The Call Agent can use the DeleteConnection command to delete an existing connection. The DeleteConnection command may also be used by a gateway to indicate that a connection can no longer be sustained.
- * The Call Agent can use the AuditEndpoint and AuditConnection commands to audit the status of an "endpoint" and any connections associated with it. Network management beyond the capabilities provided by these commands are generally desirable, e.g. information about the status of the embedded client. Such capabilities are expected to be supported by the use of the Simple Network Management Protocol (SNMP) and definition of a MIB which is outside the scope of this specification.
- * The Gateway can use the RestartInProgress command to notify the Call Agent that the gateway, or a group of endpoints managed by the gateway, is being taken out of service or is being placed back in service.

These services allow a controller (normally, the Call Agent) to instruct a gateway on the creation of connections that terminate in an "endpoint" attached to the gateway, and to be informed about events occurring at the endpoint. An endpoint may be for example:

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- * A specific trunk circuit, within a trunk group terminating in a gateway,
- * A specific announcement handled by an announcement server.

Connections are grouped into "calls". Several connections, that may or may not belong to the same call, can terminate in the same endpoint . Each connection is qualified by a "mode" parameter, which can be set to "send only", "receive only", "send/receive", "conference", "data", "inactive", "loopback", "continuity test", "network loop back" or "network continuity test."

The handling of the audio signals received on these connections is determined by the mode parameters:

- * Audio signals received in data packets through connections in "receive", "conference" or "send/receive" mode are mixed and sent to the endpoint.
- * Audio signals originating from the endpoint are transmitted over all the connections whose mode is "send", "conference" or "send/receive."
- * In addition to being sent to the endpoint, audio signals received in data packets through connections in "conference" mode are replicated to all the other connections whose mode is "conference."

The "loopback" and "continuity test" modes are used during maintenance and continuity test operations. There are two flavors of continuity test, one specified by ITU and one used in the US. In the first case, the test is a loopback test. The originating switch will send a tone (the go tone) on the bearer circuit and expect the terminating switch to loopback the circuit. If the originating switch sees the same tone returned (the return tone), the COT has passed. If not, the COT has failed. In the second case, the go and return tones are different. The originating switch sends a certain go tone. The terminating switch detects the go tone, it asserts a different return tone in the backwards direction. When the originating switch detects the return tone, the COT is passed. If the originating switch never detects the return tone, the COT has failed.

If the mode is set to "loopback", the gateway is expected to return the incoming signal from the endpoint back into that same endpoint. This procedure will be used, typically, for testing the continuity of trunk circuits according to the ITU specifications.

If the mode is set to "continuity test", the gateway is informed that the other end of the circuit has initiated a continuity test procedure

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according to the GR specification. The gateway will place the circuit in the transponder mode required for dual-tone continuity tests.

If the mode is set to "network loopback", the audio signals received from the connection will be echoed back on the same connection.

If the mode is set to "network continuity test", the gateway will process the packets received from the conenction according to the transponder mode required for dual-tone continuity test, and send the processed signal back on the connection.

2.3.1. NotificationRequest

The NotificationRequest commands are used to request the gateway to send notifications upon the occurrence of specified events in an endpoint. For example, a notification may be requested for when a gateway detects that an endpoint is receiving tones associated with fax communication. The entity receiving this notification may decide to use a different type of encoding method in the connections bound to this endpoint.

> NotificationRequest(EndpointId, NotifiedEntity, RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling, DetectEvents)

EndpointId is the name for the endpoint in the gateway where NotificationRequest executes, as defined in section 2.1.1.

NotifiedEntity is an optional parameter that specifies where the notifications should be sent. When this parameter is absent, the notifications should be sent to the originator of the NotificationRequest.

RequestIdentifier is used to correlate this request with the notifications that it triggers.

RequestedEvents is a list of events that the gateway is requested to detect and report. Such events include, for example, fax tones, continuity tones, or on-hook transition. To each event is associated an action, which can be:

* Notify the event immediately, together with the accumulated list of observed events,

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- * Swap audio,
- Accumulate the event in an event buffer, but don't notify yet,
- * Accumulate according to Digit Map,
- * Keep Signal(s) active,
- * process the Embedded Notification Request,
- * Ignore the event.

Some actions can be combined. In particular:

- * The "swap audio" action can be combined with "Notify", "Accumulate" and "Ignore."
- * The "keep signal active" action can be combined with "Notify", "Accumulate", "Accumulate acording to Digit Map" and "Embedded Notification Request."
- * The "Embedded Notification Request" can be combined with "Accumulate." It can also be combined with Notify, if the gateway is allowed to issue several Notify commands in response to a single Notification request.

Events that are not specified in the list will, by default, be ignored.

The Swap Audio action can be used when a gateway handles more than one active connection on an endpoint. This will be the case for three-way calling, call waiting, and possibly other feature scenarios. In order to avoid the round-trip to the Call Agent when just changing which connection is attached to the audio functions of the endpoint, the NotificationRequest can map an event (usually hook flash, but could be some other event) to a local function swap audio, which selects the "next" connection in a round robin fashion. If there is only one connection, this action is effectively a no-op.

If signal(s) are desired to start when an event being looked for occurs, the "Embedded NotificationRequest" action can be used. The embedded NotificationRequest may include a new list of RequestedEvents, SignalRequests and a new digit map as well. The semantics of the embedded NotificationRequest is as if a new NotificationRequest was just received with the same NotifiedEntity, and RequestIdentifier. The quarantine buffer will not be cleared (see later).

MGCP implementations shall be able to support at least one level of embedding. An embedded NotificationRequest that respects this

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limitation shall not contain another Embedded NotificationRequest.

DigitMap is an optional parameter that allows the Call Agent to provision the gateways with a digit map according to which digits will be accumulated. If this optional parameter is absent, the previously defined value is retained. This parameter must be defined, either explicitly or through a previous command, if the RequestedEvent parameters contain an request to "accumulate according to the digit map." The collection of these digits will result in a digit string. The digit string is initialized to a null string upon reception of the NotificationRequest, so that a subsequent notification only returns the digits that were collected after this request.

SignalRequests is a parameter that contains the set of signals that the gateway is asked to apply to the endpoint , such as, for example ringing, or continuity tones. Signals are identified by their name, which is an event name, and may be qualified by parameters.

The action triggered by the SignalRequests is synchronized with the collection of events specified in the RequestedEvents parameter. For example, if the NotificationRequest mandates "ringing" and the event request ask to look for an "off-hook" event, the ringing shall stop as soon as the gateway detect an off hook event. The formal definition is that the generation of all "Time Out" signals shall stop as soon as one of the requested events is detected, unless the "Keep signals active" action is associated to the specified event.

The specific definition of actions that are requested via these Signal-Requests, such as the duration of and frequency of a DTMF digit, is outside the scope of MGCP. This definition may vary from location to location and hence from gateway to gateway.

The RequestedEvents and SignalRequests refer to the same events. In one case, the gateway is asked to detect the occurrence of the event, and in the other case it is asked to generate it. The specific events and signals that a given endpoint can detect or perform are determined by the list of event packages that are supported by that end point. Each package specifies a list of events and actions that can be detected or performed. A gateway that is requested to detect or perform an event belonging to a package that is not supported by the specify endpoint shall return an error. When the event name is not qualified by a package name, the default package name for the end point is assumed. If the event name is not registered in this default package, the gateway shall return an error.

The Call Agent can send a NotificationRequest whose requested signal list is empty. It will do so for example when tone generation should stop.

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The optional QuarantineHandling parameter specifies a set of handling options:

- * whether the quarantined events should be processed or discarded (the default is to discard them.)
- * whether the gateway is expected to generate at most one notification (step by step), or multiple notifications (loop), in response to this request (the default is exactly one.)

When the parameter is absent, the default value is assumed.

DetectEvents is an optional parameter that specifies a list of events that the gateway is requested to detect during the quarantine period. When this parameter is absent, the events that should be detected in the quarantine period are those listed in the request list, including those for which the "ignore" action is specified.

Some events and signals, such as the in-line ringback or the quality alert, are performed or detected on connections terminating in the end point rather than on the endpoint itself. The following rules apply:

- * when the notification request ask for a signal to be sent "from" the endpoint, as is the case for example of inline ringback, then the corresponding audio signals should be sent over all the connections that would otherwise receive audio signals from the endpoint.
- * when the notification request ask for an event to be observed on incoming connections, then the endpoint would try to detect the event on all the active connections. The following notify command will specify the identifier of the connection on which the signal was observed.

2.3.2. Notifications

Notifications are sent via the Notify command and are sent by the gateway when the observed events occur.

> Notify(EndpointId, NotifiedEntity, RequestIdentifier, ObservedEvents)

EndpointId is the name for the endpoint in the gateway which is issuing the Notify command, as defined in section 2.1.1. The identifier should be a fully qualified endpoint name, including the domain name of the gateway. The local part of the name shall not use the wildcard

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convention.

NotifiedEntity is an optional parameter that identifies the entity to which the notifications is sent. This parameter is equal to the NotifiedEntity parameter of the NotificationRequest that triggered this notification. The parameter is absent if there was no such parameter in the triggering request. In this case, the notification is sent to the entity from which the request was received.

RequestIdentifier is parameter that repeats the RequestIdentifier parameter of the NotificationRequest that triggered this notification. It is used to correlate this notification with the request that triggered it.

ObservedEvents is a list of events that the gateway detected. A single notification may report a list of events that will be reported in the order in which they were detected. The list may only contain the identification of events that were requested in the RequestedEvents parameter of the triggering NotificationRequest. It will contain the events that were either accumulated (but not notified) or treated according to digit map (but no match yet), and the final event that triggered the detection or provided a final match in the digit map.

2.3.3. CreateConnection

This command is used to create a connection.

ConnectionId, [SpecificEndPointId,] [LocalConnectionDescriptor] <--- CreateConnection(CallId, EndpointId, NotifiedEntity, LocalConnectionOptions, Mode, RemoteConnectionDescriptor, RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling, DetectEvents)

This function is used to create a connection between two endpoints. A connection is defined by its endpoints. The input parameters in CreateConnection provide the data necessary to build a gateway's "view" of a connection.

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CallId is a globally unique parameter that identifies the call (or session) to which this connection belongs. This parameter is unique within the whole network of gateways; connections that belong to the same call share the same call-id. The call-id can be used to identify calls for reporting and accounting purposes.

EndpointId is the identifier for the connection endpoint in the gateway where CreateConnection executes. The EndpointId can be fully-specified by assigning a value to the parameter EndpointId in the function call or it may be under-specified by using the "anyone" wildcard convention. If the endpoint is underspecified, the endpoint name will be assigned by the gateway and its complete value returned in the SpecificEndPointId parameter of the response.

The NotifiedEntity is an optional parameter that specifies where the Notify or DeleteConnection commands should be sent. If the parameter is absent, the Notify or DeleteConnection commands should be sent to the originator of the CreateConnection command.

LocalConnectionOptions is a parameter used by the Call Agent to direct the handling of the connection by the gateway. The fields contained in LocalConnectionOptions are the following:

- * Encoding Method,
- Packetization period,
- * Bandwidth,
- * Type of Service,
- * Usage of echo cancellation,
- Usage of silence suppression or voice activity detection,
- * Usage of signal level adaptation and noise level reduction, or "gain control."
- * Usage of reservation service,
- * Usage of RTP security,
- * Type of network used to carry the connection.

The values of several of these fields are defined in the SDP standard. For each of the first three fields, the Call Agent has three options:

* It may state exactly one value, which the gateway will then use for

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the connection,

- * It may provide a loose specification, such as a list of allowed encoding methods or a range of packetization periods,
- * It may simply provide a bandwidth indication, leaving the choice of encoding method and packetization period to the gateway.

The bandwidth specification shall not contradict the specification of encoding methods and packetization period. If an encoding method is specified, then the gateway is authorized to use it, even if it results in the usage of a larger bandwidth than specified.

The LocalConnectionOptions parameter may be absent in the case of a data call.

The Type of Service specifies the class of service that will be used for the connection. When the connection is transmitted over an IP network, the parameters encodes the 8-bit type of service value parameter of the IP header. When the Type of Service is not specified, the gateway shall use a default or configured value.

The gateways can be instructed to perform a reservation, for example using RSVP, on a given connection. When a reservation is needed, the call agent will specify the reservation profile that should be used, which is either "controlled load" or "guaranteed service." The absence of reservation can be indicated by asking for the "best effort" service, which is the default value of this parameter. When reservation has been asked on a connection, the gateway will:

- * start emitting RSVP "PATH" messages if the connection is in "sendonly", "send-receive", "conference", "network loop back" or "network continuity test" mode,
- * start emitting RSVP "RESV" messages as soon as it receives "PATH"
 messages if the connection is in "receive-only", "send-receive",
 "conference", "network loop back" or "network continuity test"
 mode.

The RSVP filters will be deduced from the characteristics of the connection. The RSVP resource profiles will be deduced from the connection's bandwidth and packetization period.

By default, the telephony gateways always perform echo cancellation. However, it is necessary, for some calls, to turn off these operations. The echo cancellation parameter can have two values, "on" (when the echo cancellation is requested) and "off" (when it is turned off.)

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The telephony gateways may perform gain control, in order to adapt the level of the signal. However, it is necessary, for example for modem calls, to turn off this function. The gain control parameter may either be specified as "automatic", or as an explicit number of decibels of gain. The default is to not perform gain control, which is equivalent to specifying a gain of 0 decibels.

The telephony gateways may perform voice activity detection, and avoid sending packets during periods of silence. However, it is necessary, for example for modem calls, to turn off this detection. The silence suppression parameter can have two values, "on" (when the detection is requested) and "off" (when it is turned off.) The default is "off."

The Call agent can request the gateway to enable encryption of the audio Packets. It does so by providing an key specification, as specified in RFC 2327. By default, encryption is not used.

The Call Agent may instruct the gateway to prepare the connection on a specified type of network. The type of network is encoded as in the "connection-field" parameter of the SDP standard. Possible values are IN (Internet), ATM and LOCAL. The parameter is optional; if absent, the network is determined by the type of gateway.

RemoteConnectionDescriptor is the connection descriptor for the remote side of a connection, on the other side of the IP network. It includes the same fields as in the LocalConnectionDescriptor, i.e. the fields that describe a session according to the SDP standard. This parameter may have a null value when the information for the remote end is not known yet. This occurs because the entity that builds a connection starts by sending a CreateConnection to one of the two gateways involved in it. For the first CreateConnection issued, there is no information available about the other side of the connection. This information may be provided later via a ModifyConnection call. In the case of data connections (mode=data), this parameter describes the characteristics of the data connection.

Mode indicates the mode of operation for this side of the connection. The mode are "send", "receive", "send/receive", "conference", "data", "inactive", "loopback", "continuity test", "network loop back" or "network continuity test." The expected handling of these modes is specified in the introduction of the "Gateway Handling Function" section. Some end points may not be capable of supporting all modes. If the command specifies a mode that the endpoint cannot support, and error shall be returned.

The gateway returns a ConnectionId, that uniquely identifies the connection within one endpoint , and a LocalConnectionDescriptor, which is a session description that contains information about addresses and RTP

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ports, as defined in SDP. The LocalConnectionDescriptor is not returned in the case of data connections. The SpecificEndPointId is an optional parameter that identifies the responding endpoint. It can be used when the EndpointId argument referred to a "any of" wildcard name. When a SpecificEndPointId is returned, the Call Agent should use it as the EndpointId value is successive commands referring to this call.

After receiving a "CreateConnection" request that did not include a RemoteConnectionDescriptor parameter, a gateway is in an ambiguous situation. Because it has exported a LocalConnectionDescriptor parameter, it can potentially receive packets. Because it has not yet received the RemoteConnectionDescriptor parameter of the other gateway, it does not know whether the packets that it receives have been authorized by the Call Agent. It must thus navigate between two risks, i.e. clipping some important announcements or listening to insane data. The behavior of the gateway is determined by the value of the Mode parameter:

- * If the mode was set to ReceiveOnly, the gateway should accept the voice signals and transmit them through the endpoint.
- * If the mode was set to Inactive, Loopback, Continuity Test, the gateway should refuse the voice signals.
- * If the mode was set to Network Loopback or Network Continuity Test, the gateway should perform the expected echo or Response.

Note that the mode values SendReceive, Conference, Data and SendOnly don't make sense in this situation. They should be treated as ReceiveOnly and Inactive.

The RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling and DetectEvents parameters are optional. They can be used by the Call Agent to transmit a NotificationRequest that is executed simultaneously with the creation of the connection. For example, when the Call Agent wants to initiate a call to an residential gateway, it should:

- * ask the residential gateway to prepare a connection, in order to be sure that the user can start speaking as soon as the phone goes off hook,
- * ask the residential gateway to start ringing,
- * ask the residential gateway to notify the Call Agent when the phone goes off-hook.

This can be accomplished in a single CreateConnection command, by also transmitting the RequestedEvent parameters for the off hook event, and

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the SignalRequest parameter for the ringing signal.

When these parameters are present, the creation and the NotificationRequests should be synchronized, which means that both should be accepted, or both refused. In our example, the CreateConnection may be refused if the gateway does not have sufficient resources, or cannot get adequate resources from the local network access, and the off-hook Notification-Request can be refused in the glare condition, if the user is already off-hook. In this example, the phone should not ring if the connection cannot be established, and the connection should not be established if the user is already off hook.

The NotifiedEntity parameter, if present, applies to both the CreateConnection and the NotificationRequest command.

2.3.4. ModifyConnection

This command is used to modify the characteristics of a gateway's "view" of a connection. This "view" of the call includes both the local connection descriptors as well as the remote connection descriptor.

[LocalConnectionDescriptor] <--- ModifyConnection(CallId,

EndpointId, ConnectionId, NotifiedEntity, LocalConnectionOptions, Mode, RemoteConnectionDescriptor, RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling, DetectEvents)

The parameters used are the same as in the CreateConnection command, with the addition of a ConnectionId that identifies the connection within the call. This parameter is returned by the CreateConnection function, as part of the local connection descriptor. It uniquely identifies the connection within the context of the endpoint.

The EndpointId should be a fully qualified endpoint name. The local name shall not use the wildcard convention.

The ModifyConnection command can be used to affect parameters of a connection in the following ways:

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- * Provide information on the other end of the connection, through the RemoteConnectionDescriptor.
- * Activate or deactivate the connection, by changing the value of the Mode parameter. This can occur at any time during the connection, with arbitrary parameter values.
- * Change the sending parameters of the connection, for example by switching to a different coding scheme, changing the packetization period, or modifying the handling of echo cancellation.

Connections can only be activated if the RemoteConnectionDescriptor has been provided to the gateway. The receive only mode, however, can be activated without the provision of this descriptor.

The command will only return a LocalConnectionDescriptor if the local connection parameters, such as RTP ports, were modified. (Usage of this feature is actually for further study.)

The RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling and DetectEvents parameters are optional. They can be used by the Call Agent to transmit a NotificationRequest that is executed simultaneously with the modification of the connection. For example, when a connection is accepted, the calling gateway should be instructed to place the circuit in send-receive mode and to stop providing ringing tones.

This can be accomplished in a single ModifyConnection command, by also transmitting the RequestedEvent parameters, for the on hook event, and an empty SignalRequest parameter, to stop the provision of ringing tones.

When these parameters are present, the modification and the NotificationRequests should be synchronized, which means that both should be accepted, or both refused. The NotifiedEntity parameter, if present, applies to both the ModifyConnection and the NotificationRequest command.

2.3.5. DeleteConnection (from the Call Agent)

This command is used to terminate a connection. As a side effect, it collects statistics on the execution of the connection.

Connection-parameters <-- DeleteConnection(CallId, EndpointId, ConnectionId, NotifiedEntity, RequestedEvents,

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RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling, DetectEvents)

The endpoint identifier, in this form of the DeleteConnection command, shall be fully qualified. Wildcard conventions shall not be used.

In the general case where a connection has two ends, this command has to be sent to both gateways involved in the connection. Some connections, however, may use IP multicast. In this case, they can be deleted individually.

After the connection has been deleted, the endpoint should be placed in inactive mode. Any loopback that has been requested for the connection should be cancelled.

In response to the DeleteConnection command, the gateway returns a list of parameters that describe the status of the connection. These parameters are:

Number of packets sent:

The total number of RTP data packets transmitted by the sender since starting transmission on this connection. The count is not reset if the sender changes its synchronization source identifier (SSRC, as defined in RTP), for example as a result of a Modify command. The value is zero if the connection was set in "receive only" mode.

Number of octets sent:

The total number of payload octets (i.e., not including header or padding) transmitted in RTP data packets by the sender since starting transmission on this connection. The count is not reset if the sender changes its SSRC identifier, for example as a result of a ModifyConnection command. The value is zero if the connection was set in "receive only" mode.

Number of packets received:

The total number of RTP data packets received by the sender since starting reception on this connection. The count includes packets received from different SSRC, if the sender used several values. The value is zero if the connection was set in "send only" mode.

Number of octets received:

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The total number of payload octets (i.e., not including header or padding) transmitted in RTP data packets by the sender since starting transmission on this connection. The count includes packets received from different SSRC, if the sender used several values. The value is zero if the connection was set in "send only" mode.

Number of packets lost:

The total number of RTP data packets that have been lost since the beginning of reception. This number is defined to be the number of packets expected less the number of packets actually received, where the number of packets received includes any which are late or duplicates. The count includes packets received from different SSRC, if the sender used several values. Thus packets that arrive late are not counted as lost, and the loss may be negative if there are duplicates. The count includes packets received from different SSRC, if the sender used several values. The number of packets expected is defined to be the extended last sequence number received, as defined next, less the initial sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence number received. The count includes packets received from different sequence is zero if the connection was set in "send only" mode. This parameter is omitted if the connection was set in "data" mode.

Interarrival jitter:

An estimate of the statistical variance of the RTP data packet interarrival time measured in milliseconds and expressed as an unsigned integer. The interarrival jitter J is defined to be the mean deviation (smoothed absolute value) of the difference D in packet spacing at the receiver compared to the sender for a pair of packets. Detailed computation algorithms are found in RFC 1889. The count includes packets received from different SSRC, if the sender used several values. The value is zero if the connection was set in "send only" mode. This parameter is omitted if the connection was set in "data" mode.

Average transmission delay:

An estimate of the network latency, expressed in milliseconds. This is the average value of the difference between the NTP timestamp indicated by the senders of the RTCP messages and the NTP timestamp of the receivers, measured when this messages are received. The average is obtained by summing all the estimates, then dividing by the number of RTCP messages that have been received. This parameter is omitted if the connection was set in "data" mode.

For a detailed definition of these variables, refer to RFC 1889.

When the connection was set up over an ATM network, the meaning of these

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parameters may change: Number of packets sent: The total number of ATM cells transmitted since starting transmission on this connection. Number of octets sent: The total number of payload octets transmitted in ATM cells. Number of packets received: The total number of ATM cells received since starting reception on this connection. Number of octets received: The total number of payload octets received in ATM cells. Number of packets lost: Should be determined as the number of cell losts, or set to zero if the adaptation layer does not enable the gateway to assess losses. Interarrival jitter: Should be understood as the interarrival jitter between ATM cells. Average transmission delay: The gateway may not be able to assess this parameter over an ATM network. It could simply report a null value. When the connection was set up over an LOCAL interconnect, the meaning of these parameters is defined as follow: Number of packets sent: Not significant. Number of octets sent: The total number of payload octets transmitted over the local connection. Number of packets received: Not significant. Number of octets received: The total number of payload octets received over the connection. Number of packets lost: Not significant. A value of zero is assumed. Interarrival jitter: Not significant. A value of zero is assumed.

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Average transmission delay: Not significant. A value of zero is assumed.

The NotifiedEntity, RequestedEvents, RequestIdentifier, DigitMap, SignalRequests, QuarantineHandling and DetectEvents parameters are optional. They can be used by the Call Agent to transmit a Notification-Request that is executed simultaneously with the deletion of the connection. For example, when a user hangs up is accepted, the gateway should be instructed to delete the connection and to start looking for an off hook event.

This can be accomplished in a single DeleteConnection command, by also transmitting the RequestedEvent parameters, for the off hook event, and an empty SignalRequest parameter.

When these parameters are present, the DeleteConnection and the NotificationRequests should be synchronized, which means that both should be accepted, or both refused.

2.3.6. DeleteConnection (from the VoIP gateway)

In some circumstances, a gateway may have to clear a connection, for example because it has lost the resource associated with the connection, or because it has detected that the endpoint no longer is capable or willing to send or receive voice. The gateway terminates the connection by using a variant of the DeleteConnection command:

> DeleteConnection(CallId, EndpointId, ConnectionId, Reason-code, Connection-parameters)

In addition to the call, endpoint and connection identifiers, the gateway will also send the call's parameters that would have been returned to the Call Agent in response to a DeleteConnection command. The reason code indicates the cause of the disconnection.

2.3.7. DeleteConnection (multiple connections, from the Call Agent)

A variation of the DeleteConnection function can be used by the Call Agent to delete multiple connections at the same time. The command can be used to delete all connections that relate to a Call for an endpoint:

> DeleteConnection(CallId, EndpointId)

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It can also be used to delete all connections that terminate in a given endpoint:

DeleteConnection(EndpointId)

Finally, Call Agents can take advantage of the hierarchical naming structure of endoints to delete all the connections that belong to a group of endpoints. In this case, the "local name" component of the EndpointID will be specified using the "all value" wildcarding convention. The "any value" convention shall not be used. For example, if endpoints names are structured as the combination of a physical interface name and a circuit number, as in "X35V3+A4/13", the Call Agent may replace the circuit number by a wild card character "*", as in "X35V3+A4/*". This "wildcard" command instructs the gateway to delete all the connections that where attached to circuits connected to the physical interface "X35V3+A4".

After the connections have been deleted, the endpoint should be placed in inactive mode. Any loopback that has been requested for the connections should be cancelled.

This command does not return any individual statistics or call parameters.

2.3.8. Audit Endpoint

The AuditEndPoint command can be used by the Call Agent to find out the status of a given endpoint.

The EndpointId identifies the endpoint that is being audited. The "all of" wildcard convention can be used to start auditing of a group of endpoints. If this convention is used, the gateway should return the list

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of endpoint identifiers that match the wildcard in the EndPointIdList parameter. It shall not return any parameter specific to one of these endpoints.

When a non-wildcard EndpointId is specified, the (possibly empty) RequestedInfo parameter describes the information that is requested for the EndpointId specified. The following endpoint info can be audited with this command:

RequestedEvents, DigitMap, SignalRequests, RequestIdentifier, NotifiedEntity, ConnectionIdentifiers, DetectEvents, and Capabilities.

The response will in turn include information about each of the items for which auditing info was requested:

- * RequestedEvents, the list of events that the endpoint is currently looking for.
- * DigitMap, the digit map the endpoint is currently using.
- * SignalRequests, the list of the signals that are currently being applied to the endpoint
- * RequestIdentifier, the RequestIdentifier for the last Notification-Request received by this endpoint (includes NotificationRequest embedded in Connection handling primitives).
- * NotifiedEntity, the current notified entity for any active NotificationRequest (including embedded NotificationRequest).
- * ConnectionIdentifiers, the list of ConnectionIdentifiers for all connections that currently exist for the endpoint specified.
- * DetectEvents, the list of events that are currently detected in quarantine mode.
- * Capabilities requests a list of LocalConnectionOptions, the parameter values such as compression algorithms, packetization period, connection networks that the gateway is ready to support for that endpoint, as defined in the "CreateConnection" section. In addition, the option can also be used to encode the event packages that the endpoint supports, and the list of connection modes that the gateway is ready to support for that endpoint.

The Call Agent may then decide to use the AuditConnection command to obtain further information about the connections.

If no info was requested and the EndpointId refers to a valid endpoint,

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the gateway simply returns a positive acknowledgement.

If no NotifiedEntity has been specified in the last NotificationRequest, the notified entity defaults to the source address of the last NotificationRequest command received for this connection.

2.3.9. Audit Connection

The AuditConnection command can be used by the Call Agent to retrieve the parameters attached to a connection:

The EndpointId parameter specifies the endpoint that handles the connection. The wildcard conventions shall not be used.

The ConnectionId parameter is the identifier of the audited connection, within the context of the specified endpoint.

The (possibly empty) RequestedInfo describes the information that is requested for the ConnectionId within the EndpointId specified. The following connection info can be audited with this command:

CallId, NotifiedEntity, LocalConnectionOptions, Mode, RemoteConnectionDescriptor, LocalConnectionDescriptor, ConnectionParameters

The AuditConnectionResponse will in turn include information about each of the items auditing info was requested for:

- * CallId, the CallId for the call the connection belongs to.
- * NotifiedEntity, the current notified entity for the Connection.
- * LocalConnectionOptions, the LocalConnectionOptions that was supplied for the connection.
- * Mode, the current mode of the connection.

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- * RemoteConnectionDescriptor, the RemoteConnectionDescriptor that was supplied to the gateway for the connection.
- * LocalConnectionDescriptor, the LocalConnectionDescriptor the gateway supplied for the connection.
- * ConnectionParameters, the current value of the connection parameters for the connection.

If no info was requested and the EndpointId is valid, the gateway simply checks that the connection exists, and if so returns a positive ack-nowledgement.

If no NotifiedEntity has been specified for the connection, the notified entity defaults to the source address of the last connection handling command received for this connection.

2.3.10. Restart in progress

The RestartInProgress command is used by the gateway to signal that An endpoint, or a group of endpoint, is taken in or out of service.

The EndPointId identifies the endpoint that are taken in or out of service. The "all of" wildcard convention may be used to apply the command to a group of endpoint, such as for example all endpoints that are attached to a specified interface, or even all endpoints that are attached to a given gateway. The "any of" wildcard convention shall not be used.

The RestartMethod parameter specified the type of restart. Three values have been defined:

- * A "graceful" restart method indicates that the specified endpoints will Be taken out of service after the specified delay. The established connections are not yet affected, but the Call Agent should refrain to establish new connections, and should try to gracefully tear down the existing connections.
- * A "forced" restart method indicates that the specified endpoints are taken abruptely out of service. The established connections, if any, are lost.

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* A "restart" method indicates that service will be restored on the endpoints after the specified "restart delay." There are no connections that are currently established on the endpoints.

The optional "restart delay" parameter is expressed as a number of seconds. If the number is absent, the delay value should be considered null. In the case of the "graceful" method, a null delay indicates that the call agent should simply wait for the natural termination of the existing connections, without establishing new connections. The restart delay is always considered null in the case of the "forced" method.

2.4. Race conditions

In order to implement proper call signalling, the Call Agent must keep track of the state of the endpoint, and the gateway must make sure that events are properly notified to the call agent. MGCP deals with race conditions through the notion of a "quarantine list" and through explicit detections of desynchronizations.

MGCP does not assume that the transport mechanism will maintain the order of comamnd and responses. This may cause race conditions, that may be obviated through a proper behavior of the call agent.

In some cases, many gateways may decide to restart operation at the same time. This may occur, for example, if an area loses power or trasnmission capability during an earthquake or an ice storm. When power and transmission are reestablished, many gateways may decide to send "RestartInProgress" commands in synchronicity, leading to very unstable operation.

2.4.1. Quarantine list

MGCP controlled gateways will receive "notification requests" that ask them to watch for a list of "events." The two protocol elements that determine the handling of these events are the "Requested Events" list and the "Digit Map."

When the endpoint is initialized, the requested events list and the digit map are empty. After reception of a command, the gateway starts observing the endpoint for occurences of the event mentioned in the list.

The events are examined as they occur. The action that follows is determined by the "action" parameter associated to the event in the list of requested events, and also by the digit map. The events that are defined as "accumulate" are accumulated in a list of events, the events that are marked as treated according to the digit map will be accumulated in the dialed string. This will go on until one event is

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encountered that triggers a Notification to the "notified entity."

The gateway, at this point, will transmit the notification command and will place the endpoint in a "notification" state. As long as the endpoint is in this notification state, the events that may occur on the endpoint are accumulated, but not transmitted, even if they had normally triggered a notification. The events are, in a sense, "quarantined." All events that are specified in the DetectEventList, or, in the absence of that list, all events that are refered to in the RequestEventList, should be detected and quarantined, regardless of the action associated to the event.

The endpoint exits the notification state when the acknowledgement of the notification command is received. During that period, the notification command may be retransmitted, as specified in section 3.5.

Following that point, the behavior of the gateway depends on the value of The QuarantineHandling parameter in the notification request. If the Call Agent specified that it expected at most one notification in response to the notification request command, then the gateway should simply keep on accumulating events in the quarantine list until it receives the next notification request command.

If the gateway is authorized to send multiple successive notifications, It will proceed as follow. When the gateway receives the acknowledgement, it resets the "current dial string" of the endpoint to a null value and starts processing the list of quarantined events, using the already received list of requested events and digit map. When processing these events, the gateway may encounter an event which requires a Notify command to be sent. If that is the case, the gateway can adopt one of the two following behaviors:

- * it can immediately transmit a Notify command that will report all events that where accumulated until the triggering event, included, leaving the unprocessed events in the quarantine list,
- * or it can attempt to empty the quarantined list and transmit a single Notify command reporting several sets of events and possibly several dial strings. The dial string is reset to a null value after each triggering event that satisfies the DigitMap. The events that follow the last triggering event are left in the quarantine list.

If the gateway transmits a Notify command, the end point will remain in the notification state until the acknowledgement is received. If the gateway does not find a quarantined event that requests a notification, it places the end point in a normal state. Events are then processed as they come, in exactly the same way as if a Notification Request command

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had just been received.

A gateway may receive at any time a new Notification Request command for the end point. When a new notification request is received in the notification state, the endpoint is taken out of the notification state without waiting for the acknowledgement of the notification command, and the pending notification command will not be repeated.

After receiving the Notification Request command, the requested events list and digit map are replaced by the newly received parameters, and the accumulated dial string is reset to a null value. The behavior is conditioned by the value of the QuarantineHandling parameter. The parameter may specify that quarantine events should be discarded, in which case they will be. If the parameter specifies that the event should be processed, the gateway will start processing the list of quarantined events, using the newly received list of requested events and digit map. When processing these events, the gateway may encounter an event which requires a Notify command to be sent. If that is the case, the gateway will immediately transmit a Notify command that will report all events that where accumulated until the triggering event, included, leaving the unprocessed events in the quarantine list, and will enter the Notification state.

2.4.2. Explicit detection

A key element of the state of several endpoints is the position of the hook. A race condition may occur when the user decides to go off-hook before the Call Agent has the time to ask the gateway to notify an off hook event (the "glare" condition well known in telephony), or if the user goes on-hook before the Call Agent has the time to request the event's notification.

To avoid this race condition, the gateway should check the condition of the endpoint before acknowledging a NotificationRequest. It should return an error:

- 1- If the gateway is requested to notify an "off hook" transition while the phone is already off hook,
- 2- If the gateway is requested to notify an "on hook" or "flash hook" condition while the phone is already on hook.

The other state variables of the gateway, such as the list of RequestedEvent or list of requested signals, are entirely replaced after each successful NotificationRequest, which prevents any long term discrepancy between the Call Agent and the gateway.

When a NotificationRequest is unsuccessful, the list of RequestedEvents

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and requested signals are emptied. They must be reinstated by a new request.

Another race condition may occur when a notification is issued shortly before the reception by the gateway of a NotificationRequest. The RequestIdentifier is used to correlate Notify commands with NotificationRequest commands.

2.4.3. Ordering of commands, and treatment of disorder

MGCP does not mandate that the underlying transport protocol guarantees the sequencing of commands send to a gateway or an endpoint. This property tends to maximize the timeliness of actions, but it has a few draw backs. For example:

- * Notification commands may be delayed and arrive to the call agent after the transmission of a new Notification Request command,
- * If a new NotificationRequest is transmitted before a previous one is acknowledged, there is no guarantee that the previous one will not be received in second position.

Call Agents that want to guarantee consistent operation of the end points can use the following rules:

- 1) When a gateway handles several endpoints, commands pertaining to the different endpoints can be sent in parallel, for example following a model where each endpoint is controlled by its own process or its own thread.
- 2) When several connections are created on the same endpoint, commands pertaining to different connections can be sent in parallel.
- 3) On a given connection, there should normally be only one outstanding command (create or modify). However, a DeleteConnection command can be issued at any time. In consequence, a gateway may sometime receive a ModifyConnection command that applies to a previously deleted connection. Such commands should be ignored, and an error code should be returned.
- 4) On a given endpoint, there should normally be only one outstanding NotificationRequest command at any time. The RequestId parameter should be used to correlate notification commands with the triggering notification request.
- 5) In some cases, a "global" DeleteConnection command that applies to a group of endpoints can step in front of a pending CreateConnection command. The Call Agent should individually delete all

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connections whose completion was pending at the time of the global DeleteConnection command.

These rules do not affect the gateway, which should always respond to commands.

2.4.4. Fighting the restart avalanche

Let's suppose that a large number of gateways are powered on simultaneously. If they were to all initiate a RestartInProgress transaction, the call agent would very likely be swamped, leading to message losses and network congestion during the critical period of service restoration. In order to prevent such avalanches, the following behavior is suggested:

- When a gateway is powered on, it should initiate a restart timer to a random value, uniformly distributed between 0 and a maximum waiting delay (MWD). Care should be taken to avoid synchronicity of the random number generation between multiple gateways that would use the same algorithm.
- 2) The gateway should then wait for either the end of this timer, the reception of a command from the call agent, or the detection of a local user activity, such as for example an off-hook transition on a residential gateway.
- 3) When the timer elapses, when a command is received, or when an activity is detected, the gateway should initiate the restart procedure.

The value of MWD is a configuration parameter that depends on the type of the gateway. The following]reasoning can be used to determine the value of this delay on residential gateways.

Call agents are typically dimensioned to handle the peak hour traffic load, during which, in average, 10% of the lines will be busy, placing calls whose average duration is typically 3 minutes. The processing of a call typically involves 5 to 6 MGCP transactions between each end point and the call agent. This simple calculation shows that the call agent is expected to handle 5 to 6 transactions for each end point, every 30 minutes on average, or, to put it otherwise, about one transaction per end point every 5 to 6 minutes on average. This suggest that a reasonable value of MWD for a residential gateway would be 10 to 12 minutes. In the absence of explicit configuration, residential gateways should adopt a value of 360 seconds for MWD.

The same reasoning suggests that the value of MWD should be much shorter for trunking gateways or for business gateways, because they handle a

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large number of endpoints, and also because the usage rate of these endpoints is much higher than 10% during the peak busy hour, a typical value being 60%. These endpoints, during the peak hour, are this expected to contribute about one transaction per minute to the call agent load. A reasonable algorithm is to make the value of MWD per "trunk" endpoint six times shorter than the MWD per residential gateway, and also inversely proportional to the number of endpoints that are being restarted. for example MWD should be set to 2.5 seconds for a gateway that handles a T1 line, or to 60 milliseconds for a gateway that handles a T3 line.

2.5. Return codes and error codes.

All MGCP commands are acknowledged. The acknowledgment carries a return code, which indicates the status of the command. The return code is an integer number, for which three ranges of values have been defined:

- values between 200 and 299 indicate a successful completion,
- * values between 400 and 499 indicate a transient error,
- * values between 500 and 599 indicate a permanent error.

The values that have been already defined are listed in the following table:

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Code	Meaning
200	The requested transaction was executed normally.
250	The connection was deleted.
400	The transaction could not be executed,
	due to a transient error.
401	The phone is already off hook
402	The phone is already on hook
500	The transaction could not be executed,
	because the endpoint is unknown.
501	The transaction could not be executed,
	because the endpoint is not ready.
502	The transaction could not be executed,
	because the endpoint does not have
	sufficient resources
510	The transaction could not be executed,
	because a protocol error was detected.
511	The transaction could not be executed,
	because the command contained an
	unrecognized extension.
512	The transaction could not be executed,
	because the gateway is not equipped to
	detect one of the requested events.
513	The transaction could not be executed,
	because the gateway is not equipped to
	generate one of the requested signals.
514	The transaction could not be executed,
	because the gateway cannot send the
-1-	specified announcement.
515	The transaction refers to an incorrect
F1C	connection-id (may have been already deleted)
510	The transaction refers to an unknown call-1d.
51/	Unsupported or invalid mode.
518	Onsupported or unknown package.
513	The transportion gould not be evenued
5 <u>7</u> 0	bogouge the ordpoint is "restarting"
	because the enupoint is restarting .

3. Media Gateway Control Protocol

The MGCP implements the media gateway control interface as a set of transactions. The transactions are composed of a command and a mandatory response. There are eight types of command:

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- * CreateConnection
- * ModifyConnection
- * DeleteConnection
- * NotificationRequest
- * Notify
- * AuditEndpoint
- * AuditConnection
- * RestartInProgress

The first four commands are sent by the Call Agent to a gateway. The Notify command is sent by the gateway to the Call Agent. The gateway may also send a DeleteConnection as defined in 2.3.6. The Call Agent may send either of the Audit commands to the gateway. The Gateway may send a RestartInProgress command to the Call Agent.

3.1. General description

All commands are composed of a Command header, optionally followed by a session description.

All responses are composed of a Response header, optionally followed by a session description.

Headers and session descriptions are encoded as a set of text lines, separated by a line feed character. The headers are separated from the session description by an empty line.

MGCP uses a transaction identifier to correlate commands and responses. The transaction identifier is encoded as a component of the command header and repeated as a component of the response header (see section 3.2.1, 3.2.1.2 and 3.3).

Transaction identifiers have values between 1 and 9999999999. An MGCP entity shall not reuse a transaction identifier sooner than 3 minutes after completion of the previous command in which the identifier was used.

3.2. Command Header

The command header is composed of:

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draft-huitema-MGCP-v0r1-01.txt

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- * A command line, identifying the requested action or verb, the endpoint towards which the action is requested, and the MGCP protocol version,
- * A set of parameter lines, composed of a parameter name followed by a parameter value.

3.2.1. Command line

The command line is composed of:

- * The name of the requested verb,
- * The identification of the transaction,
- * The name of the endpoint that should execute the command (in notifications, the name of the endpoint that is issuing the notification),
- * The protocol version.

These four items are encoded as strings of printable ASCII characters, separated by white spaces, i.e. the ASCII space (0x20) or tabulation (0x09) characters. It is recommended to use exactly one ASCII space separator.

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3.2.1.1. Coding of the requested verb

The verbs that can be requested are encoded as four letter upper or lower case ASCII codes (comparisons should be case insensitive) as defined in the following table:

Verb	Code
CreateConnection	CRCX
ModifyConnection	MDCX
DeleteConnection	DLCX
NotificationRequest	RQNT
Notify	NTFY
AuditEndpoint	AUEP
AuditConnection	AUCX
RestartInProgress	RSIP

The transaction identifier is encoded as a string of up to 9 decimal digits. In the command lines, it immediately follows the coding of the verb.

New verbs may be defined in further versions of the protocol. It may be necessary, for experimentation purposes, to use new verbs before they are sanctioned in a published version of this protocol. Experimental verbs should be identified by a four letter code starting with the letter X, such as for example XPER.

3.2.1.2. Coding of the endpoint names

The endpoint names are encoded as e-mail addresses, as defined in RFC 821. In these addresses, the domain name identifies the system where the endpoint is attached, while the left side identifies a specific endpoint on that system.

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Examples of such addresses can be:

hrd4/56@gw23.example.net	Circuit number 56 in interface "hrd4" of the Gateway 23
	of the "Example" network
Call-agent@ca.example.net	Call Agent for the "example" network
Busy-signal@ann12.example.net	The "busy signal" virtual endpoint in the announcement server number 12.

The name of notified entities is expressed with the same syntax, with the possible addition of a port number as in:

Call-agent@ca.example.net:5234

3.2.1.3. Coding of the protocol version

The protocol version is coded as the key word MGCP followed by a white space and the version number, and optionally followed by a profile name.. The version number is composed of a major version, coded by a decimal number, a dot, and a minor version number, coded as a decimal number. The version described in this document is version 1.2.

The profile name, if present, is represented by a string of letters and digits. Profile names may be defined for user communities who want to apply restrictions or other profiling to MGCP.

In the initial messages, the version will be coded as:

MGCP 0.1

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3.2.2. Parameter lines

Parameter lines are composed of a parameter name, which in most cases is composed of a single upper case character, followed by a colon, a white space and the parameter value. The parameter that can be present in commands are defined in the following table:

Parameter name	Code	Parameter value				
CallId	C	Hexadecimal string,				
	Í	at most 32 characters				
ConnectionId	I	Hexadecimal string,				
	ĺ	at most 32 characters				
NotifiedEntity	N	An identifier, in RFC 821 format,				
		composed of an arbitrary string and				
		of the domain name of the requesting				
		entity, possibly completed by a port				
		number, as in:				
		Call-agent@ca.example.net:5234				
RequestIdentifier	X	Hexadecimal string,				
		at most 32 characters				
LocalConnectionOptions	L	See				
description						
Connection Mode	М	See description				
RequestedEvents	R	See description				
SignalRequests	S	See description				
DigitMap	D	A text encoding of a digit map				
ObservedEvents	0	See description				
ConnectionParameters	P	See description				
ReasonCode	E	An arbitrary character string				
SpecificEndpointID	Z	An identifier, in RFC 821 format,				
		composed of an arbitrary string,				
		followed by an "@" followed by the				
		domain name of the gateway to which				
		this endpoint is attached.				
RequestedInto	F. 1	See description				
QuarantineHandiing		See description				
		See Description				
RestartMethod		See description				
KestartDelay		A number of seconds, encoded as				

The parameters are not necessarily present in all commands. The following table provides the association between parameters and commands. The letter M stands for mandatory, O for optional and F for forbidden.

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Parameter name	CR	MD	DL	RQ	NT	AU	AU	RS
	CX	CX	CX	NT	FY	ED	CX	IP
CallId	М	М	0	F	F	F	F	F
ConnectionId	F	М	0	F	F	F	M	F
RequestIdentifier	0	0	0	М	М	F	F	F
LocalConnection	0	0	F	F	F	F	F	F
Options								
Connection Mode	М	М	F	F	F	F	F	F
RequestedEvents	0	0	0	0*	F	F	F	F
SignalRequests	0	0	0	0*	F	F	F	F
NotifiedEntity	0	0	0	0	0	F	F	F
ReasonCode	F	F	0	F	F	F	F	F
ObservedEvents	F	F	F	F	М	F	F	F
DigitMap	0	0	0	0	F	F	F	F
Connection	F	F	0	F	F	F	F	F
parameters								
Specific Endpoint ID	F	F	F	F	F	F	F	F
RequestedInfo	F	F	F	F	F	М	М	F
QuarantineHandling	0	0	0	0	0	F	F	F
DetectEvents	0	0	0	0	0	F	F	F
RestartMethod	F	F	F	F	F	F	F	M
RestartDelay	F	F	F	F	F	F	F	0
RemoteConnection	0	0	F	F	F	F	F	F
Descriptor								

Note (*) that the RequestedEvents and SignalRequests parameters are optional in the NotificationRequest. If these parameters are omitted, the corresponding lists will be considered empty.

If implementers need to experiment with new parameters, for example when developing a new application of MGCP, they should identify these parameters by names that start with the string "X-" or "X+", such as for example:

X-FlowerOfTheDay: Daisy

Parameter names that start with "X+" are mandatory parameter extensions. An MGCP entity that receives a mandatory parameter extension that it cannot understand should refuse to execute the command. It should respond with an error code 511 (Unrecognized extension).

Parameter names that start with "X-" are non critical parameter

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extensions. An MGCP entity that receives a mandatory parameter extension that it cannot understand can safely ignore that parameter.

3.2.2.1. Local connection options

The local connection options describe the operational parameters that the Call Agent suggests to the gateway. These parameters are:

- * The packetization period in milliseconds, encoded as the keyword "p", followed by a colon and a decimal number. If the Call Agent specifies a range of values, the range will be specified as two decimal numbers separated by an hyphen.
- * The preferred type of compression algorithm, encoded as the keyword "a", followed by a character string. If the Call Agent specifies a list of values, these values will be separated by a semicolon.
- * The bandwidth in kilobits per second (1000 bits per second), encoded as the keyword "b", followed by a colon and a decimal number. If the Call Agent specifies a range of values, the range will be specified as two decimal numbers separated by an hyphen.
- * The echo cancellation parameter, encoded as the keyword "e", followed by a colon and the value "on" or "off".
- * The gain control parameter, encoded as the keyword "gc", followed by a colon a value which can be either the keyword "auto" or a decimal number (positive or negative) representing the number of decibels of gain.
- * The silence suppression parameter, encoded as the keyword "s", followed by a colon and the value "on" or "off".
- * The type of service parameter, encoded as the keyword "t", followed by a colon and the value encoded as two hexadecimal digits.
- * The resource reservation parameter, encoded as the keyword "r", followed by a colon and the value "g" (guaranteed service), "cl" (controlled load) or "be" (best effort).
- * The encryption key, encoded as the keyword "k" followed by a colon and a key specification, as defined for the parameter "K" of SDP (RFC 2327).
- * The type of network, encoded as the keyword "nt" followed by a colon and the type of network encoded as the keyword "IN", "ATM" or "LOCAL".

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Each of the parameters is optional. When several parameters are present, the values are separated by a comma.

Examples of connection descriptors are:

L: p:10, a:G.711 L: p:10, a:G.711;G.726-32 L: p:10-20, b: 64 L: b:32-64, e:off

LocalConnectionOptions may furthermore be used by the embedded client to inform the call agent about its capabilities when audited. In that case, parameters will have conventional values that are related to capabilities rather than actual connections, and may also contain a list of supported packages, and a list of supported modes:

*

A list of supported codecs. The following parameters will apply to all codecs specified in this list. If there is a need to specify that some parameters, such as e.g. silence suppression, are only compatible with some codecs, then the gateway will return several LocalConnectionOptions parameters, one for each set of codecs.

Packetization Period:

A range may be specified.

Bandwidth:

A range corresponding to the range for packetization periods may be specified (assuming no silence suppression). If absent, the values will be deduced from the codec type.

Echo Cancellation:

"on" if echo cancellation is supported for this codec, "off" otherwise. The default is support.

Silence Suppression:

"on" if silence suppression is supported for this codec, "off" otherwise. The default is support.

Gain Control:

"0" if gain control is not supported. The default is support.

Type of Service:

The value "0" indicates no support for type of service, all other values indicate support for type of service. The default is support.

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Resource Reservation:

The parameter indicates the reservation services that are supported, in addition to best effort. The value "g" is encoded when the gateway supports both the guaranteed and the controlled load service, "cl" when only the controlled load service is supported. The default is "best effort."

Encryption Key:

Encoding any value indicates support for encryption. Default is no support.

Type of network: The keyword "nt" followed by a co

The keyword "nt", followed by a colon and a semicolon separated list of supported network types. This parameter is optional.

Event Packages

The event packages supported by this endpoint encoded as the keyword "v", followed by a colon and a character string. If a list of values is specified, these values will be separated by a semicolon. The first value specified will be the default package for that endpoint.

Modes

A comma-separated list of supported connection modes for this endpoint.

3.2.2.2. Connection parameters

Connection parameters are encoded as a string of type and value pairs, where the type is a two letter identifier of the parameter, and the value a decimal integer. Types are separated from value by an '=' sign. Parameters are encoded from each other by a comma.

The connection parameter types are specified in the following table:

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Connection parameter	Code 	Connection parameter value
Packets sent	PS	The number of packets that were sent on the connection.
Octets sent	OS	The number of octets that were sent on the connection.
Packets received	PR	The number of packets that were received on the connection.
Octets received	OR	The number of octets that were received on the connection.
Packets lost	PL 	The number of packets that were not received on the connection, as deduced from gaps in the sequence number.
Jitter	JI	The average inter-packet arrival jitter, in milliseconds, expressed as an integer number
Latency	LA 	Average latency, in milliseconds, expressed as an integer number.

An example of connection parameter encoding is:

P: PS=1245, OS=62345, PR=0, OR=0, PL=0, JI=0, LA=48

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3.2.2.3. Connection mode

The connection mode describes the mode of operation of the connection. The possible values are:

Mode	Meaning
M: sendonly	The gateway should only send packets
M: recvonly	The gateway should only receive packets
M: sendrecv	The gateway should send
	and receive packets
M: confrnce	The gateway should place
	the connection in conference mode
M: inactive	The gateway should neither
	send nor receive packets
M: loopback	The gateway should place
	the circuit in loopback mode.
M: conttest	The gateway should place
	the circuit in test mode.
M: netwloop	The gateway should place
	the connection in network loopback mode.
M: netwtest	The gateway should place
	the connection in network
M. data	The gateway should use the sirguit
M. Uala	for network access for data
	(a g DDD SLTD atg)
	(c.g., rrr, Dhr, ccc.).

3.2.2.4. Coding of event names

Event names are composed of an optional package name, separated by a slash (/) from the name of the actual event. Event names are used in the RequestedEvents, SignalRequests and ObservedEvents parameter. The following are valid examples of event names:

L/hu	on-hook transition, in the line package
F/0	digit 0 in the MF package
fh	Flash-hook, assuming that the line package
	is a default package for the end point.

In addition, the range and wildcard notation of events can be used, instead of individual names, in the RequestedEvents and DetectEvents

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parameters. The following are valid examples of such notations:

M/[0-9]	Digits 0 to 9 in the MF package
fh	Flash-hook, assuming that the line package
	is a default package for the end point.
[0-9*#A-D]	All digits and letters in the DTMF
	packages (default for endpoint).
Т/\$	All events in the trunk packages.

3.2.2.5. RequestedEvents

The RequestedEvent parameter provides the list of events that have been requested. The event codes are described in the previous section.

Each event can be qualified by a requested action, or by a list of actions. The actions, when specified, are encoded as a list of keywords, enclosed in parenthesis and separated by commas. The codes for the various actions are:

Action	Code
Notify immediately	N
Accumulate	A
Treat according to digit map	D
Swap	S
Ignore	I
Keep Signal(s) active	K
Embedded Notification Request	Е

When no action is specified, the default action is to notify the event. This means that, for example, ft and ft(N) are equivalent. Events that are not listed are ignored.

The digit-map action can only be specified for the digits, letters and interdigit timers in the MF and DTMF packages.

The requested list is encoded on a single line, with event/action groups separated by commas. Examples of RequestedEvents encoding are:

R: hu(N), hf(S,N) R: hu(N), [0-9#T](D)

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In the case of the "enable" action, the embedded notification request parameters are encoded as a list of three parameter groups, delimitated by an opening and closing parenthesis, and separated by semicolons.

R: hd(E([0-9#T](D),hu(N);dl;[0-9].[#T]))

The first parameter group is the enabled value of the RequestedEvents parameter, followed by the enabled value of the SignalRequests parameter, optionally followed by the enabled value of the DigitMap. This last parameter is optional. If it is absent, the current value should be used. The following are valid examples of embedded requests:

```
R: hd(E([0-9#T](D),hu(N);dl;[0-9].[#T]))
R: hd(E([0-9#T](D),hu(N);dl)
```

3.2.2.6. SignalRequests

The SignalRequests parameter provides the name of the signals that have been requested. Each signal is identified by a name, as indicated in the previous section.

Several signals, such as for example announcement or ADSI display, can be qualified by additional parameters:

- * the name and parameters of the announcement,
- * the string that should be displayed.

These parameters will be encoded as a set of UTF8 character strings, spearated by comams and enclosed within parenthesis, as in:

- S: adsi("123456 Francois Gerard")
- S: ann(no-such-number, 1234567)

When several signals are requested, their codes are separated by a comma, as in:

S: asdi(123456 Your friend), rg

3.2.2.7. ObservedEvent

The observed event parameters provides the list of events that have been observed. The event codes are the same as those used in the NotificationRequest. Events that have been accumulated according to the digit map are grouped in a single string. Examples of observed actions are:

0: hu

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O: 8295555T O: hf, hf, hu

The packet arrival event is used to notify that at least one packet was recently sent to an Internet address that is observed by an endpoint. The event report includes the Internet address, in standard ASCII encoding, between parenthesis:

0: pa(192.96.41.1)

The call back event is used to notify that a call back has been requested during the initial phase of a data connection. The event report includes the identification of the user that should be called back, between parenthesis:

O: cbk(user25)

3.2.2.8. RequestedInfo

The RequestedInfo parameter contains a comma separated list of parameter codes, as defined in the "Parameter lines" section. For example, if one wants to audit the value of the NotifiedEntity, RequestIdentifier, RequestedEvents, SignalRequests, DigitMap, QuarantineHandling and DetectEvents parameters, The value of the RequestedInfo parameter will be:

F:N,X,R,S,D,Q,T

The capabilities request, in the AuditEndPoint command, is encoded by the keyword "A", as in:

F:A

3.2.2.9. QuarantineHandling

The quarantine handling parameter contains a list of comma separated keywords:

- * The keyword "process" or "discard" to indicate the treatment of quarantined events. If neither process or discard is present, discard is assumed.
- * The keyword "step" or "loop" to indicate whether exactly at most one notification is expected, or whether multiple notifications are allowed. If neither step or loop is present, step is assumed. The

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following values are valid examples:

Q:loop Q:process Q:discard,loop

3.2.2.10. DetectEvents

The DetectEvent parameter is encoded as a comma separated list of events, such as for example:

T: hu,hd,hf,[0-9#*]

3.2.2.11. RestartMethod

The RestartMethod parameter is encoded as one of the keywords "grace-ful", "forced" or "restart", as for example:

RM:restart

3.3. Format of response headers

The response header is composed of a response line, optionally followed by headers that encode the response parameters.

The response line starts with the response code, which is a three digit numeric value. The code is followed by a white space, the transaction identifier, and an optional commentary.

The following table describe the parameters whose presence is mandatory or optional in a response header, as a function of the command that triggered the response. The letter M stands for mandatory, O for optional and F for forbidden.

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Parameter nameParameter name	CR	MD	DL	RQ	NT	AU	AU	RS
	CX	CX	CX	NT	FY	ED	CX	IP
CallId	F	F	F	F	F	F	0	F
ConnectionId	0	F	F	F	F	F	0	F
RequestIdentifier	F	F	F	F	F	0	F	F
LocalConnection	F	F	F	F	F	0	0	F
Options								
Connection Mode	F	F	F	F	F	F	0	F
RequestedEvents	F	F	F	F	F	0	F	F
SignalRequests	F	F	F	F	F	0	F	F
NotifiedEntity	F	F	F	F	F	F	F	0
ReasonCode	F	F	F	F	F	F	F	F
ObservedEvents	F	F	F	F	F	0	F	F
DigitMap	F	F	F	F	F	0	F	F
Connection	F	F	0	F	F	F	0	F
parameters								Í
Specific Endpoint ID	0	F	F	F	F	F	F	F
RequestedInfo	F	F	F	F	F	0	F	F
QuarantineHandling	F	F	F	F	F	0	F	F
DetectEvents	F	F	F	F	F	0	F	F
RestartMethod	F	F	F	F	F	F	F	F
RestartDelay	F	F	F	F	F	F	F	F
LocalConnection	М	0	F	F	F	F	0*	F
Descriptor								
RemoteConnection	F	F	F	F	F	F	0*	F
Descriptor							Í	ĺ

In the case of a CreateConnection message, the response line is followed by a Connection-Id parameter. It may also be followed a Specific-Endpoint-Id parameter, if the creation request was sent to a wildcarded Endpoint-Id.

In the case of a DeleteConnection message, the response line is followed by a Connection Parameters parameter, as defined in section 3.2.2.2.

A LocalConnectionDescriptor should be transmitted with a positive response (code 200) to a CreateConnection. It may be transmitted in response to a ModifyConnection command, if the modification resulted in a modification of the session parameters. The LocalConnectionDescriptor is encoded as a "session description," as defined in section 3.4. It is separated from the response header by an empty line.

The current syntax does not allow to carry both a local connection

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descriptor and a remote connection descriptor in the same response.

3.4. Encoding of the session description

The session description is encoded in conformance with the session description protocol, SDP. MGCP implementations are expected to be fully capable of parsing any conformant SDP message, and should send session descriptions that strictly conform to the SDP standard. The usage of SDP actually depends on the type of session that is being, as specified in the "mode" parameter:

- * if the mode is set to "data", the session description describes the configuration of a data access service.
- * if the mode is set to any other value, the session description is for an audio service.

For an audio service, the gateway will consider the information provided in SDP for the "audio" media. For a data service, the gateway will consider the information provided for the "network-access" media.

3.4.1. Usage of SDP for an audio service

In a telephony gateway, we only have to describe sessions that use exactly one media, audio. The parameters of SDP that are relevant for the telephony application are:

At the session description level:

* The IP address of the remote gateway (in commands) or of the local gateway (in responses), or multicast address of the audio conference, encoded as an SDP "connection data" parameter. This parameter specifies the IP address that will be used to exchange RTP packets.

For the audio media:

- Media description field (m) specifying the audio media, the transport port used for receiving RTP packets by the remote gateway (commands) or by the local gateway (responses), the RTP/AVP transport, and the list of formats that the gateway will accept. This list should normally always include the code 0 (reserved for G.711).
- Optionally, RTPMAP attributes that define the encoding of dynamic audio formats,
- * Optionally, a packetization period (packet time) attribute (Ptime) defining the duration of the packet,

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- * Optionally, an attribute defining the type of connection (sendonly, recvonly, sendrecv, inactive)
- * The IP address of the remote gateway (in commands) or of the local gateway (in responses), if it is not present at the session level.

An example of SDP specification for an audio connection could be:

v=0
c=IN IP4 128.96.41.1
m=audio 3456 RTP/AVP 0 96
a=rtpmap:96 G726-32/8000

There is a request, in some environments, to use the MGCP to negotiate connections that will use other transmission channels than RTP over UDP and IP. This will be detailed in an extension to this document.

3.4.2. Usage of SDP in a network access service

The parameters of SDP that are relevant for a data network access application are:

For the data media:

- Media description field (m) specifying the network access media, identified by the code "m=nas/xxxx", where "xxxx" describes the access control method that should be used for parametrizing the network access, as specified below. The field may also specify the port that should be used for contacting the server, as specified in the SDP syntax.
- * Connection address parameter (c=) specifying the address, or the domain name, of the server that implement the access control method. This parameter may also be specified at the session level.
- * Optionally, a bearer type attribute (a=bearer:) describing the type of data connection to be used, including the modem type.
- * Optionally, a framing type attribue (a=framing:) describing the type of framing that will be used on the channel.
- * Optionally, attributes describing the called number (a=dialed:), the number to which the call was delivered (a=called:) and the calling number (a=dialing:).
- * Optionally, attributes describing the range of addresses that could be used by the dialup client on its LAN (a=subnet:).

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* Optionally, an encryption key, encoded as specified in the SDP protocol (k=).

The connection address shall be encoded as specified in the SDP standard. It will be used in conjunction with the port specified in the media line to access a server, whose type will one of:

Method name	Method description
radius	Authentication according
	to the Radius protocol.
tacacs	Authentication according
	to the TACACS+ protocol.
diameter	Authentication according
	to the Diameter protocol.
l2tp	Level 2 tunneling protocol.
_	The address and port are those of the LNS.
login	Local login. (There is normally
	no server for that method.)
none	No authentication required.
j	(The call was probably vetted
İ	by the Call Agent.)
Í	

If needed, the gateway may use the key specified in the announcement to access the service. That key, in particular, may be used for the establishment of an L2TP tunnel.

The bearer attribute is composed of a bearer name and an optional extension. The bearer type specifies the type of modulation (modem name) or, in the case of digital connections, the type of ISDN service (8 bits, 7 bits). When an extension is present, it is separated from the bearer name by a single slash (/). The valid values of the bearer attribute are defined in the following table:

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Type of bearer description	Example of values
ITU modem standard ITU modem standard qualified by a manufacturer name	V.32, V.34, V.90. v.90/3com, v.90/rockwell, v.90/xxx
Well known modem types ISDN transparent access, 64 kbps ISDN64 + V.110 ISDN64 + V.120 ISDN transparent access, 56 kbps Informal identification	X2, K56flex ISDN64 ISDN64/V.110 ISDN64/V.120 ISDN56 (Requires coordination between the Call Agent and the gateway)

The valid values of the framing attribute are defined in the following table:

Type of framing description	Example of values
PPP, asynchronous framing	ppp-asynch
PPP, HDLC framing	ppp-hdlc
SLIP, asynchronous	slip
Asynchronous, no framing	asynch

The network access authentication parameter provides instructions on the access control that should be exercized for the data call. This optional attribute is encoded as:

Where the parameters "network type", "address type", and "connection address" are formatted as defined for the connection address parameter (c=) in SDP, and where the "prefix length" is a decimal representation of the number of bits in the prefix.

Examples of SDP announcement for the network access service could be:

v=0
m=nas/radius
c=IN IP4 radius.example.net

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```
a=bearer:v.34
a=framing:ppp-asynch
a=dialed:18001234567
a=called:12345678901
a=dialing:12340567890
v=0
m=nas/none
c=IN IP4 128.96.41.1
a=subnet:IN IP4 123.45.67.64/26
a=bearer:isdn64
a=framing:ppp-sync
a=dialed:18001234567
a=dialing:2345678901
v=0
c=IN IP4 access.example.net
```

```
m=nas/l2tp
k=clear:some-shared-secret
a=bearer:v.32
a=framing:ppp-asynch
a=dialed:18001234567
a=dialing:2345678901
```

3.4.3. Usage of SDP for ATM connections

The specification of the SDP payload for ATM connections will be described in a companion document, "Usage of MGCP to control Voice over ATM gateways." The following text is indicative.

The SDP payload will specify:

- * That the connection is to be established over an ATM interface, using the "c=" parameter of SDP to specify an address in the ATM family, the ATM addressing variant (NSAP, UNI, E.164) and the ATM address.
- * The "m=audio" parameter will specify the audio encoding and, if needed, the VPI and VCI.
- * Additional attributes parameters (a=) will be used to specify the ATM coding variants, such as the type of adaptation layer and the error correction or loss compennesation algorithms.

An example of SDP payload for an ATM connection could be:

v=0

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c=ATM NSAP 47.0091.8100.0000.0060.3e64.fd01.0060.3e64.fd01.fe m=audio 5/1002 ATM/AVP G.711u a=connection_type:AAL2

3.4.4. Usage of SDP for local connections

When MGCP is used to set up internal connections within a single gateway, the SDP format is used to encode the parameters of that connection. The following parameters will be used:

- * The connection parameter (C=) will specify that the connection is local, using the keyword "LOCAL" as network type space, the keyword "EPN" (endpoint name) as address type, and the name of the endpoint as the connection-address.
- * The "m=audio" parameter will specify a port number, which will always be set to 0, the type of protocol, always set to the keyword LOCAL, and the type of encoding, using the same conventions used for RTP (RTP payload numbers.) The type of encoding should normally be set to 0 (G.711).

An example of local SDP payload could be:

v=0
c=LOCAL EPN X35V3+A4/13
m=audio 0 LOCAL 0

3.5. Transmission over UDP

MGCP messages are transmitted over UDP. Commands are sent to one of the IP addresses defined in the DNS for the specified endpoint . The responses are sent back to the source address of the commands.

When no port is specified for the endpoint, the commands should be sent to the default MGCP port, 2427.

MGCP messages, being carried over UDP, may be subject to losses. In the absence of a timely response, commands are repeated. MGCP entities are expected to keep in memory a list of the responses that they sent to recent transactions, i.e. a list of all the responses they sent over the last 30 seconds, and a list of the transactions that are currently being executed. The transaction identifiers of incoming commands are compared to the transaction identifiers of the recent responses. If a match is found, the MGCP entity does not execute the transaction, but simply repeats the response. The remaining commands will be compared to the list of current transaction. If a match is found, the MGCP entity does

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not execute the transaction, which is simply ignored.

It is the responsibility of the requesting entity to provide suitable time outs for all outstanding commands, and to retry commands when time outs have been exceeded. Furthermore, when repeated commands fail to be acknowledged, it is the responsibility of the

requesting entity to seek redundant services and/or clear existing or pending connections.

The specification purposely avoids specifying any value for the retransmission timers. These values are typically network dependent. The retransmission timers should normally estimate the timer by measuring the time spent between the sending of a command and the return of a response. One possibility is to use the algorithm implemented in TCP-IP, which uses two variables:

- * the average acknowledgement delay, AAD, estimated through an exponentially smoothed average of the observed delays,
- * the average deviation, ADEV, estimated through an exponentially smoothed average of the absolute value of the difference between the observed delay and the current average

The retransmission timer, in TCP, is set to the sum of the average delay plus N times the average deviation.

After the any retransmission, the MGCP entity should do the following:

- * It should double the estimated value of the average delay, AAD
- * It should compute a random value, uniformly distributed between 0.5 AAD and AAD
- * It should set the retransmission timer to the sum of that random value and N times the average deviation.

This procedure has two effects. Because it includes an exponentially increasing component, it will automatically slow down the stream of messages in case of congestion. Because it includes a random component, it will break the potential synchronization between notifications triggered by the same external event.

3.5.1. Piggy backing

There are cases when a Call Agent will want to send several messages at the same time to the same gateways. When several MGCP messages have to be sent in the same UDP packets, they should be separated by a line of

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text that contain a single dot, as in for example:

200 2005 OK . DLCX 1244 card23/21@trgw-7.example.net MGCP 0.1 C: A3C47F21456789F0 I: FDE234C8

The piggy-backed messages should be processed exactly has if they had been received in several simultaneous messages.

4. Security requirements

If unauthorized entities could use the MGCP, they would be able to setup unauthorized calls, or to interfere with authorized calls. We expect that MGCP messages will always be carried over secure Internet connections, as defined in the IP security architecture as defined in RFC 1825, using either the IP Authentication Header, defined in RFC 1826, or the IP Encapsulating Security Payload, defined in RFC 1827. The complete MGCP protocol stack would thus include the following layers:

MGCP		
UDP		
IP security (authentication or encryption)		
IP		
transmission media		

Adequate protection of the connections will be achieved if the gateways and the Call Agents only accept messages for which IP security provided an authentication service. An encryption service will provide additional protection against eavesdropping, thus forbidding third parties from monitoring the connections set up by a given endpoint

The encryption service will also be requested if the session descriptions are used to carry session keys, as defined in SDP.

These procedure do not necessarily protect against denial of service attacks by misbehaving gateways or misbehaving call agents. However, they will provide an identification of these misbehaving entities, which

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should then be deprived of their authorization through maintenance procedures.

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5. Event packages and end point types

This section provides an initial definition of packages and event names. More packages can be defined in additional documents.

5.1. Basic packages

The list of basic packages includes the following:

Package	name	
Generic Media Package	G	
DTMF package	D	
MF Package	М	
Trunk Package	Т	
Line Package	L	
Handset Package	н	
RTP Package	R	
Netwark Access Server Package	N	
Announcement Server Package	A	
Script Package	Script	
	İ	

- In the tables of events for each package, there are five columns: Symbol: the unique symbol used for the event Definition: a short description of the event
 - R: an x appears in this column is the event can be Requested by the call agent.
 - S: if nothing appears in this column for an event, then the event cannot be signaled on command by the call agent. Otherwise, the following symbols identify the type of event:
 - 00 On/Off signal. The signal is turned on until commanded by the call agent to turn it off, or until the event is superseded by a new signal.
 - TO Timeout signal. The signal lasts for a given duration unless it is superseded by a new signal.
 - BR Brief signal. The event has a short, known duration.

Duration: specifies the duration of TO signals.

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5.1.1. Generic Media Package

Package Name: G

The generic media package group the events and signals that can be observed on several types of endpoints, such as trunking gateways, access gateways or residential gateways.

Symbol	Definition	R	S	Duration
 mt	Modem detected	x		
ft	Fax tone detected	x		ĺ
ld	Long duration connection	x		ĺ
pat(###)	Pattern ### detected	x	00	ĺ
rt	Ringback tone	Í	00	ĺ
cf	Confirm tone	İ	00	İ
cg	Network Congestion tone	Í	00	ĺ
it	Intercept tone	Í	00	ĺ
pt	Preemption tone	İ	00	İ
of	report failure	x		İ
		İİ		i

The pattern definition can be used for specific algorithms such as answering machine detection, tone detection, and the like.

The "long duration connection" is detected when a connection has been established for more than 1 hour.

5.1.2. DTMF package

Package name: D

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Symbol	Definition	R	S	Duration
0	DTMF 0	x	BR	l
1	DTMF 1	x	BR	
2	DTMF 2	x	BR	
3	DTMF 3	x	BR	
4	DTMF 4	x	BR	
5	DTMF 5	x	BR	
б	DTMF 6	x	BR	
7	DTMF 7	x	BR	
8	DTMF 8	x	BR	
9	DTMF 9	x	BR	
#	DTMF #	x	BR	
*	DTMF *	x	BR	
A	DTMF A	x	BR	
В	DTMF B	x	BR	
С	DTMF C	x	BR	
D	DTMF D	x	BR	
L	long duration indicator	x		2 seconds
Х	Wildcard, match	x		
	any digit 0-9			
Т	Interdigit timer	x		4 seconds
of	report failure	x		

The "interdigit timer" occurs when a delay of more than 4 seconds is observed after the end of a digit detection. The event can only be observed if the endpoint is trying to acquire digits.

The "long duration indicator" is observed when a DTMF signal is produced for a duration larger than two seconds. In this case, the gateway will detect two successive events: first, when the signal has been recognized, the DTMF signal, and then, 2 seconds later, the long duration signal.

5.1.3. MF Package

Package Name: M

Symbol	Definition	R	S	Duration
0	MF 0	 	 BR	
1	MF 1	x	BR	
2	MF 2	x	BR	
3	MF 3	x x	BR	
4	MF 4	x	BR	i
5	MF 5	x	BR	
6	MF 6	x	BR	
7	MF 7	x	BR	
8	MF 8	x	BR	
9	MF 9	x	BR	
Х	Wildcard, match	x		
	any digit 0-9	İ	İ	İ
Т	Interdigit timer	x	İ	4 seconds
К0	MF KO or KP	x	BR	İ
К1	MF K1	x	BR	i
К2	MF K2	x	BR	i
S0	MF S0 or ST	x	BR	İ
S1	MF S1	x	BR	İ
S2	MF S2	x	BR	İ
S3	MF S3	x	BR	İ
wk	Wink	x	BR	
wko	Wink off	x	BR	İ
is	Incoming seizure	x	00	
rs	Return seizure	x	00	
us	Unseize circuit	x	00	
of	report failure	x	ĺ	
				İ

The definition of the MF package events is as follow:

Wink

A transition from unseized to seized to unseized trunk states within a specified period. Typical seizure period is 100-350 msec.)

Incoming seizure Incoming indication of call attempt.

Return seizure: Seizure in response to outgoing seizure.

Unseize circuit: Unseizure of a circuit at the end of a call.

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Wink off:

A signal used in operator services trunks. A transition from seized to unseized to seized trunk states within a specified period of 100-350 ms. (To be checked)

5.1.4. Trunk Package

Package Name: T

Symbol	Definition	R	S	Duration
 col	Continuity tone (single tone, or return tone)	x	00	
co2	Continuity test (go tone, in dual tone procedures)	x	00	
lb	Loopback		00	
om	Old Milliwatt Tone (1000 Hz)	x	00	
nm	New Milliwatt Tone (1004 Hz)	x	00	
tl	Test Line	x	00	
zz	No circuit	x	00	
as	Answer Supervision	x	00	
ro	Reorder Tone	x	00	
of	report failure	х		

The definition of the trunk package signal events is as follow:

```
Continuity Tone (col):
A tone at 2010 + or - 30 Hz.
```

Continuity Test (co2): A tone at the 1780 + or - 30 Hz.

Milliwatt Tones: Old Milliwatt Tone (1000 Hz), New Milliwatt Tone (1004 Hz)

Line Test: 105 Test Line test progress tone (2225 Hz + or - 25 Hz at -10 dBm0 + or -- 0.5dB).

No circuit: (that annoying tri-tone, low to high)

Answer Supervision:

Reorder Tone:

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(120 Impulses per minute tone).

The continuity tones are used when the call agent wants to initiate a continuity test. There are two types of tests, single tone and dual tone. The Call agent is expected to know, through provisioning information, which test should be applied to a given endpoint. For example, the call agent that wants to initiate a single frequency test will send to the gateway a command of the form:

```
RQNT 1234 epx-t1/17@tgw2.example.net
X: AB123FE0
S: col
R: col
```

If it wanted instead to initiate a dual-tone test, it would send the command:

RQNT 1234 epx-t1/17@tgw2.example.net X: AB123FE0 S: co2 R: co1

The gateway would send the requested signal, and in both cases would look for the return of the 2010 Hz tone (col). When it detects that tone, it will send the corresponding notification.

The tones are of type OO: the gateway will keep sending them until it receives a new notification request.

5.1.5. Line Package

Package Name: L

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Symbol	Definition	R	S	Duration
adsi(string)	adsi display		BR	
hd	Off hook transition	x		
hu	On hook transition	x		
hf	Flash hook	x		
aw	Answer tone	x	00	
bz	Busy tone	i i	00	
wt	Call Waiting tone	i i	то	30 seconds
dl	Dial tone	i i	то	120 seconds
nbz	Network busy	x	00	
İ	(fast cycle busy)	i i		
rg	Ringing	i i	то	30 seconds
r0, r1, r2,	Distinctive ringing	i i	то	30 seconds
r3, r4, r5,		i i		
r6 or r7		i i		
p	Prompt tone	x	BR	
e	Error tone	x	BR	
sdl	Stutter dialtone	i i	BR	
v	Alerting Tone	i i	00	
y I	Recorder Warning Tone	İ	00	
sit	SIT tone	i i		
z	Calling Card Service Tone	İ	00	
oc	Report on completion	x		
ot	Off hook warning tone	į į	00	
s(###)	Distinctive tone pattern	x	BR	
of	report failure	x		ĺ

The definition of the tones is as follow:

```
Dial tone:
    A combined 350 + 440 Hz tone.
Alerting Tone:
```

```
a 440 Hz Tone of 2 second duration followed by 1/2 second of tone
every 10 seconds.
```

Recorder Warning Tone: 1400 Hz of Tone of 0.5 second duration every 15 seconds.

SIT tone: used for indicating a line is out of service.

Calling Card Service Tone: 60 ms of 941 + 1477 Hz and 940 ms of 350 + 440 Hz (dial tone),

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decaying exponentially with a time constant of 200 ms.

Distinctive tone pattern:

where ### is any number between 000 and 999, inclusive. Can be used for distinctive ringing, customized dial tone, etc.

Report on completion

The report on completion event is detected when the gateway was asked to perform one or several signals of type TO on the endpoint, and when these signals were completed without being stopped by the detection of a requested event such as off-hook transition or dialed digit. The completion report may carry as parameter the name of the signal that came to the end of its live time, as in:

O: L/oc(L/dl)

We should note that many of these definitions vary from country to country. The frequencies listed above are the one in use in North America. There is a need to accommodate different tone sets in different countries, and there is still an ongoing debate on the best way to meet that requirement:

- * One solution is to define different event packages specifying for example the German dialtone as "L-DE/DL".
- * Another solution is to use a management interface to specify on an end-point basis which frequency shall be associated to what tone.

5.1.6. Handset emulation package

Package Name: H

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Symbol	Definition	R	S	Duration
adsi(string)	 adsi display	 	 BR	
hd	Off hook transition	x	00	
hu	On hook transition	x		
hf	Flash hook	x	BR	
aw	Answer tone	x	00	
bz	Busy tone	x x		
wt	Call Waiting tone			30 seconds
d]	Dial tone $(350 + 440 \text{ Hz})$	x x		120 seconds
Inbz	Network busy			
	(fast cycle busy)			
rg	Ringing	x	то	30 seconds
r0, r1, r2,	Distinctive ringing	x	то	30 seconds
r3, r4, r5,		İ	ĺ	İ
r6 or r7		İ	İ	İ
p	Prompt tone	x	BR	
e	Error tone	x	BR	ĺ
sdl	Stutter dialtone	x	BR	İ
v	Alerting Tone	x	00	
У	Recorder Warning Tone	x	00	ĺ
t	SIT tone	x		İ
Z	Calling Card Service Tone	x	00	
oc	Report on completion	x	ĺ	
ot	Off hook warning tone	x	00	j
s(###)	Distinctive tone pattern	x	BR	j
of	report failure	x	ĺ	İ
		İ	İ	İ

The handset emulation package is an extension of the line package, to be used when the gateway is capable of emulating a handset. The difference with the line pacake is that events such as "off hook" can be signalled as well as detected.

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5.1.7. RTP Package

Package Name: R

Symbol	Definition	R	S	Duration
UC SR(###) JI(###) PL(###) qa	Used codec changed Sampling rate changed Jitter buffer size changed Packet loss exceeded Quality alert	x x x x x x x		
of	report failure	x		

Codec Changed:

Codec changed to hexadecimal codec number enclosed in parenthesis, as in UC(15), to indicate the codec was changed to PCM mu-law. Codec Numbers are specified in RFC 1890, or in a new definition of the audio profiles for RTP that replaces this RFC. Some implementations of media gateways may not allow the codec to be changed upon command from the call agent. codec changed to codec hexadecimal ##.

Sampling Rate Changed:

Sampling rate changed to decimal number in milliseconds enclosed in parenthesis, as in SR(20), to indicate the sampling rate was changed to 20 milliseconds. Some implementations of media gateways may not allow the sampling rate to be changed upon command from a call agent.

Packet Loss Exceeded:

Packet loss rate exceed the threshold of the specified decimal number of packets per 100,000 packets, where the packet loss number is contained in parenthesis. For example, PL(10) indicates packets are being dropped at a rate of 1 in 10,000 packets.

Quality alert

The packet loss rate or the comination of delay and jitter exceed a specified quality threshold.

5.1.8. Netwark Access Server Package

Package Name: N

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Symbol	Definition	R	S	Duration
pa	Packet arrival	х		ĺ
cbk	Call back request	х		
cl	Carrier lost	х		ĺ
au	Authorization succeeded	х		ĺ
ax	Authorization denied	х		ĺ
of	Report failure	х		

5.1.9. Announcement Server Package

Package Name: A

Symbol	Definition	R	S	Duration
ann(url,parms)	Play an announcement		 TO	variable
oc	Report on completion	x		İ
of	Report failure	x		
			<u> </u>	

The announcement action is qualified by an URL name and by a set of initial parameters as in for example:

S: ann(http://scripts.example.net/all-lines-busy.au)

The "operation complete" event will be detected when the announcement is played out. If the announcement cannot be played out, an operation failure event can be returned. The failure may be explained by a commentary, as in:

O: A/of(file not found)

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5.1.10. Script Package

Package Name: Script

Symbol	Definition	R	S	Duration
java(url)	Load a java script	İ	ТО	variable
perl(url)	Load a perl script		ТО	variable
tcl(url)	Load a TCL script		TO	variable
<pre>xml(url)</pre>	Load an XML script		ТО	variable
OC	Report on completion	x		
of	Report failure	x		
				Í

The "language" action define is qualified by an URL name and by a set of initial parameters as in for example:

S: script/java(http://scripts.example.net/credit-card.java,long,1234)

The current definition defines keywords for the most common languages. More languages may be defined in further version of this documents. For each language, an API specification will describe how the scripts can issue local "notificationRequest" commands, and receive the corresponding notifications.

The script produces an output which consists of one or several text string, separated by commas. The text string are reported as a commentary in the report on completion, as in for example:

O: script/oc(21223456794567,9738234567)

The failure report may also return a string, as in:

O: script/oc(21223456794567,9738234567)

The definition of the script environment and the specific actions in that environment are for further study.

5.2. Basic endpoint types and profiles

We define the following basic endpoint types and profiles:

- Trunk gateway (ISUP)
- * Trunk gateway (MF)

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- * Network Access Server (NAS)
- Combined NAS/VOIP gateway
- * Access Gateway
- * Residential Gateway
- * Announcement servers

These gateways are supposed to implement the following packages

Gateway	Supported packages
Trunk gateway (ISUP)	GM, DTMF, TK, RTP
Trunk gateway (MF)	GM, MF, DTMF, TK, RTP
Network Access Server (NAS)	GM, MF, TK, NAS
Combined NAS/VOIP gateway	GM, MF, DTMF, TK, NAS, RTP
Access Gateway (VOIP)	GM, DTMF, MF, RTP
Access Gateway (VOIP+NAS)	GM, DTMF, MF, NAS, RTP
Residential Gateway	GM, DTMF, Line, RTP
Announcement Server	ANN, RTP

Advanced announcement servers may also support the Script package.

Advanced trunking servers may support the ANN package, the Script package, and in some cases the Line and Handset package as well.

6. Versions and compatibility

MGCP version 0.1 results from the fusion of the SGCP and IPDC proposals.

6.1. Changes between MGCP and initial versions of SGCP

MGCP version 0.1 (which subsumes SGCP version 1.2) introduces the following changes from SGCP version 1.1:

- * Protocol name changed to MGCP.
- * Introduce a formal wildcarding structure in the name of endpoints, inspired from IPDC, and detailed the usage of wildcard names in each operation.
- * Naming scheme for events, introducing a package structure inspired from IPDC.

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- * New operations for audit endpoint, audit connection (requested by the Cablelabs) and restart (inspired from IPDC).
- * New parameter to control the behavior of the notification request.
- * Improved text on the detection and handling of race conditions.
- * Syntax modification for event reporting, to incorporate package names.
- * Definition of basic event packages (inspired from IPDC).
- * Incorporation of mandatory and optional extension parameters, inspired by IPDC.

SGCP version 1.1 introduces the following changes from version SGCP 1.0:

- * Extension parameters (X-??:)
- * Error Code 511 (Unrecognized extension).
- * All event codes can be used in RequestEvent, SignalRequest and ObservedEvent parameters.
- Error Code 512 (Not equipped to detect requested event).
- * Error Code 513 (Not equipped to generate requested signal).
- * Error Code 514 (Unrecognized announcement).
- Specific Endpoint-ID can be returned in creation commands.
- * Changed the code for the ASDI display from "ad" to "asdi" to avoid conflict with the digits A and D.
- * Changed the code for the answer tone from "at" to "aw" to avoid conflict with the digit A and the timer mark T
- * Changed the code for the busy tone from "bt" to "bz" to avoid conflict with the digit B and the timer mark T
- * Specified that the continuity tone value is "co" (CT was incorrectly used in several instances; CT conflicts with .)
- * Changed the code for the dial tone from "dt" to "dl" to avoid conflict with the digit D and the timer mark T
- * Added a code point for announcement requests.

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- * Added a code point for the "wink" event.
- * Set the "octet received" code in the "Connection Parameters" to "OR" (was set to RO, but then "OR" was used throughout all examples.)
- * Added a "data" mode.
- * Added a description of SDP parameters for the network access mode (NAS).
- * Added four flow diagrams for the network access mode.
- * Incorporated numerous editing suggestions to make the description easier to understand. In particular, cleared the confusion between requests, queries, functions and commands.
- * Defined the continuity test mode as specifying a dual-tone transponder, while the loopback mode can be used for a single tone test.
- * Added event code "OC", operation completed.
- * Added the specification of the "quarantine list", which clarifies the expected handling of events and notifications.
- * Added the specification of a "wildcard delete" operation.
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