

**Universal Mobile Telecommunications System (UMTS);  
Open Service Access (OSA);  
Application Programming Interface (API) Mapping for OSA;  
Part 4: Call Control Service Mapping;  
Subpart 1: API to CAP Mapping  
(3GPP TR 29.998-4-1 version 4.1.0 Release 4)**

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- z the third digit is incremented when editorial only changes have been incorporated in the document.

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## Introduction

### Structure of the OSA API Mapping (3GPP TR 29.998)

The Technical Report 3GPP TR 29.998 consists of a series of parts and subparts. An effort has been made to ensure that the part numbers used in the mapping TR correspond to the part numbers of the base OSA specification in 3GPP TS 29.198. For this reason, certain parts, for which no suitable mapping could be suggested, have not been delivered. At a later stage a mapping to a new protocol may become evident, in which case these missing parts will be developed.

The OSA documentation was defined jointly between 3GPP TSG CN WG5, ETSI SPAN 12 and the Parlay Consortium, in co-operation with the JAIN consortium. The 3GPP TR 29.998 is based on a mapping document with a wider scope, developed as part of this co-operation. Certain mappings defined in the course of this joint development are not applicable for 3GPP Release 4, which is why not all sub-parts have been delivered as part of 3GPP Release 4. However, it is expected that some will become applicable within the scope of 3GPP Release 5, which is why a common sub-part numbering is being retained, albeit with gaps for 3GPP Release 4.

If mapping for a certain Part is "Not Applicable" it can either indicate that a mapping does not exist (e.g. Part 2 Common Data), or the API is considered to be implemented directly on a physical entity, or via a proprietary mechanism.

The present document is part 4 subpart 1 of a multi-part TR covering the 3<sup>rd</sup> Generation Partnership Project: Technical Specification Group Core Network; Open Service Access (OSA); Application Programming Interface (API) Mapping for OSA, as identified below.

29.998-1: General Issues on API Mapping

29.998-2: *Not Applicable*

29.998-3: *Not Applicable*

**29.998-4-1: Call Control Service Mapping; Subpart: API to CAP Mapping**

29.998-4-2: Call Control Service Mapping; Subpart 2 generic call control INAP (not Rel4)

29.998-4-3: Call Control Service Mapping; Subpart 3 multiparty call control INAP (not Rel4)

29.998-4-4: Call Control Service Mapping; Subpart 4 multiparty call control SIP (not Rel4)

29.998-4-5: Call Control Service Mapping; Subpart 5 multimedia call control extensions mapping to SIP (not Rel4)

29.998-5-1: User Interaction Service Mapping; Subpart 1: API to CAP Mapping

29.998-5-2: User Interaction Service Mapping; Subpart 2 user interaction INAP (not Rel4)

29.998-5-3: User Interaction Service Mapping; Subpart 3 user interaction Megacop (not Rel4)

- 29.998-5-4: User Interaction Service Mapping; Subpart 4: API to SMS Mapping  
 29.998-6: User Location – User Status Service Mapping to MAP  
 29.998-7: *Not Applicable*  
 29.998-8: Data Session Control Service Mapping to CAP

OSA API specifications 29.198-family		OSA API Mapping - 29.998-family	
29.198-1	Part 1: Overview	29.998-1	Part 1: Overview
29.198-2	Part 2: Common Data Definitions	29.998-2	Not Applicable
29.198-3	Part 3: Framework	29.998-3	Not Applicable
29.198-4	Part 4: Call Control SCF	29.998-4-1	Subpart 1: Generic Call Control – CAP mapping
		29.998-4-2	
29.198-5	Part 5: User Interaction SCF	29.998-5-1	Subpart 1: User Interaction – CAP mapping
		29.998-5-2	
		29.998-5-3	
		29.998-5-4	Subpart 4: User Interaction – SMS mapping
29.198-6	Part 6: Mobility SCF	29.998-6	User Status and User Location – MAP mapping
29.198-7	Part 7: Terminal Capabilities SCF	29.998-7	Not Applicable
29.198-8	Part 8: Data Session Control SCF	29.998-8	Data Session Control – CAP mapping
29.198-9	Part 9: Generic Messaging SCF	29.998-9	Not Applicable
29.198-10	Part 10: Connectivity Manager SCF	29.998-10	Not Applicable
29.198-11	Part 11: Account Management SCF	29.998-11	Not Applicable
29.198-12	Part 12: Charging SCF	29.998-12	Not Applicable

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# 1 Scope

The present document investigates how the OSA Call Control Interface Class methods defined in 3GPP TS 29.198-4 [5] can be mapped onto CAMEL Application Part (CAP) operations and Mobile Application Part (MAP) operations. The mapping of the OSA API to the CAP and relevant MAP operations is considered informative, and not normative. An overview of the mapping TR is contained in the introduction of the present document as well as in 3GPP TR 29.998-1 [10].

The OSA specifications define an architecture that enables application developers to make use of network functionality through an open standardised interface, i.e. the OSA APIs. The API specification is contained in the 3GPP TS 29.198 series of specifications. An overview of these is available in the introduction of the present document as well as in 3GPP TS 29.198-1 [1]. The concepts and the functional architecture for the Open Service Access (OSA) are described by 3GPP TS 23.127 [3]. The requirements for OSA are defined in 3GPP TS 22.127 [2].

The present document has been defined jointly between 3GPP TSG CN WG5, ETSI SPAN 12 and the Parlay Consortium, in co-operation with the JAIN consortium.

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# 2 References

- References are either specific (identified by date of publication, edition number, version number, etc.) or non-specific.
- For a specific reference, subsequent revisions do not apply.
- For a non-specific reference, the latest version applies. In the case of a reference to a 3GPP document (including a GSM document), a non-specific reference implicitly refers to the latest version of that document *in the same Release as the present document*.

- [1] 3GPP TS 29.198-1: "Open Service Access; Application Programming Interface; Part 1: Overview".
- [2] 3GPP TS 22.127: "Stage 1 Service Requirement for the Open Service Access (OSA) (Release 4)".
- [3] 3GPP TS 23.127: "Virtual Home Environment (Release 4)".
- [4] 3GPP TR 22.905: "3GPP Vocabulary".
- [5] 3GPP TS 29.198-4: "Open Service Access; Application Programming Interface - Part 4: Call Control".
- [6] 3GPP TS 29.002: "Mobile Application Part (MAP) specification".
- [7] 3GPP TS 29.078: "CAMEL Application Part (CAP) specification – Phase 3".
- [8] 3GPP TS 22.101: "Universal Mobile Telecommunications System (UMTS): Service Aspects; Service Principles".
- [9] ITU-T Q.850: "Usage of cause and location in the Digital Subscriber Signalling System No. 1 and the Signalling System No. 7 ISDN User Part".
- [10] 3GPP TR 29.998-1: "API Mapping for Open Service Access; Part 1: General Issues on API Mapping".

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# 3 Definitions and abbreviations

## 3.1 Definitions

For the purposes of the present document, the terms and definitions given in TS 29.198-1 [1] apply.

### 3.2 Abbreviations

For the purposes of the present document, the abbreviations given in TS 29.198-1 [1] apply.

## 4 Generic Call Control CAMEL Call Flows

### 4.1 Call Control Manager

The generic call manager interface class provides the management functions to the generic call SCFs. The application programmer can use this interface to create call objects and to enable or disable call-related event notifications.

#### 4.1.1 enableCallNotification

*enableCallNotification* is used to enable call notifications to be sent to the application.

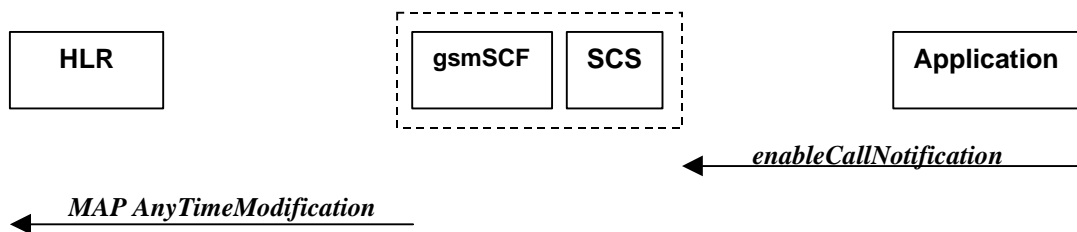


Figure 4-1: Call Flow for enableCallNotification

Table 4-1: Normal Operation

Two alternatives have been identified.

- 1 The application requests notifications to be enabled.

Pre-conditions	An agreement is established between the network operator and the service provider for the event notification to be enabled
1	The application invokes the <i>enableCallNotification</i> method
2	The gsmSCF sends a MAP <i>AnyTimeModification</i> to the HLR in order to Activate the necessary CAMEL Subscription Information (O-CSI, D-CSI, T-CSI, VT-CSI). NOTE: CAMEL phase 3 only allows for activation/deactivation of the CSI and not modification of the contents of the CSIs. The O-CSI and D-CSI will be activated if the originating address is present and the T-CSI and VT-CSI will be activated if the destination address is present.

Table 4-2: Error condition

- 2 HLR rejects CSI updates

Pre-conditions	gsmSCF had previously sent a MAP <i>AnyTimeModification</i> message to the HLR as a result of an <i>enableCallNotification</i> request from the application
1	HLR rejects the request to update the CSI
2	The gsmSCF sends an internal message to the SCS to indicate the up date failure
3	The SCS invokes the exception on <i>enableCallNotification</i>

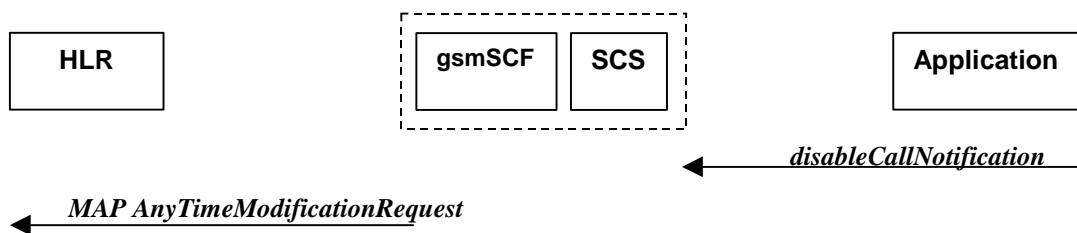


**Table 4-3: Parameter Mapping**

From: enableCallNotification	To: MAP AnyTimeModification
appInterface	
eventCriteria (TpCallEventCriteria) :	
DestinationAddress	subscriberIdentity (see Note) modificationRequestFor-CSI
OriginationAddress	subscriberIdentity (see Note) modificationRequestFor-CSI
CallEventName (TpCallEventName)	CAMEL Subscription Information - T-CSI - VT-CSI - O-CSI - D-CSI
CallNotificationType	
assignmentID	
	modificationRequestFor-SS-Info
	gsmSCF address
NOTE:	In case an address range is used, a separate MAP AnyTimeModificationRequest shall be sent for every address in the range

### 4.1.2 disableCallNotification

*disableCallNotification* is used by the application to disable call notifications.



**Figure 4-2: Call Flow for disableCallNotification**

**Table 4-4: Normal Operation**

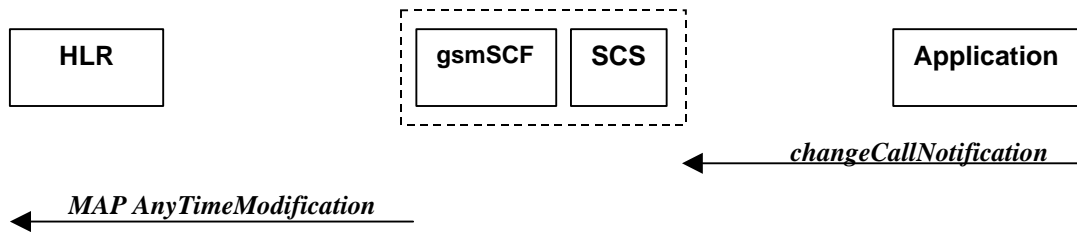
Pre-conditions	An agreement is established between the network operator and the service provider for the event notification to be disabled
1	The application invokes the <i>disableCallNotification</i> method
2	The gsmSCF sends a MAP <i>AnyTimeModification</i> to the HLR in order to de-activate the CAMEL subscription Information (O-CSI, D-CSI, T-CSI, VT-CSI). NOTE: CAMEL Phase 3 only allows the capability to activate/deactivate CSI and not to modify the triggering information. The O-CSI and D-CSI will be deactivated if the originating address is present and the T-CSI and VT-CSI will be deactivated if the destination address is present.

**Table 4-5: Parameter Mapping**

From: disableCallNotification	To: MAP AnyTimeModification
assignmentID	
	gsmSCFAddress

### 4.1.3 changeCallNotification

*changeCallNotification* is used by the application to change the call notifications previously set by *enableCallNotification()*.



**Figure 4-3: Call Flow for changeCallNotification**

**Table 4-6: Normal Operation**

Pre-conditions	Notifications have been enabled by the application.
1	The application invokes the <i>changeCallNotification</i> method
2	The gsmSCF sends a MAP <i>AnyTimeModification</i> to the HLR in order to active and de-activate the CAMEL subscription Information (O-CSI, T-CSI, VT-CSI). The SCS and gsmSCF will have to determine which CSIs to active and which to de-activate in order to reflect the changed set of notifications. The O-CSI and D-CSI will be modified if the originating address is present and the T-CSI and VT-CSI will be modified if the destination address is present

**Table 4-7: Parameter Mapping**

From: changeCallNotification	To: MAP AnyTimeModification
assignmentID	
eventCriteria (TpCallEventCriteria) :	
DestinationAddress	subscriberIdentity (see Note) modificationRequestFor-CSI
OriginationAddress	subscriberIdentity (see Note) modificationRequestFor-CSI
CallEventName (TpCallEventName)	CAMEL Subscription Information - T-CSI - VT-CSI - O-CSI - D-CSI
CallNotificationType	
	modificationRequestFor-SS-Info
	gsmSCFAddress
NOTE:	In case an address range is used, a separate MAP AnyTimeModificationRequest shall be sent for every address in the range

### 4.1.4 getCriteria

*getCriteria* is used by the application to query the event criteria set with *enableCallNotification*.

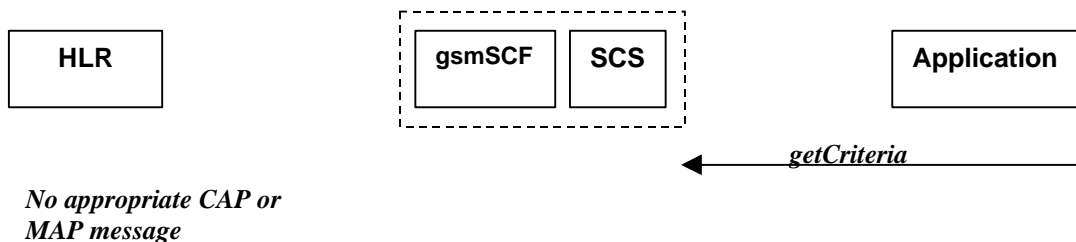


Figure 4-4: Call Flow for getCriteria

Table 4-8: Normal Operation

Pre-conditions	Notifications have been enabled by the application.
1	The application invokes the <i>getCriteria</i> method
2	The SCS returns the criteria

#### Parameter Mapping

None.

### 4.1.5 setCallLoadControl

*setCallLoadControl* is a method used to control the number of invoked methods i.e. to restrict the load placed on the application server.

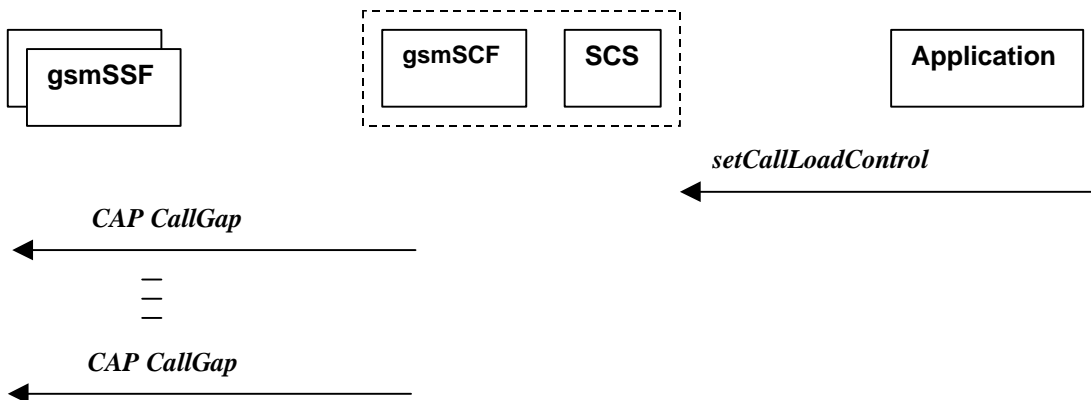


Figure 4-5: Call Flow for release

Table 4-9: Normal Operation

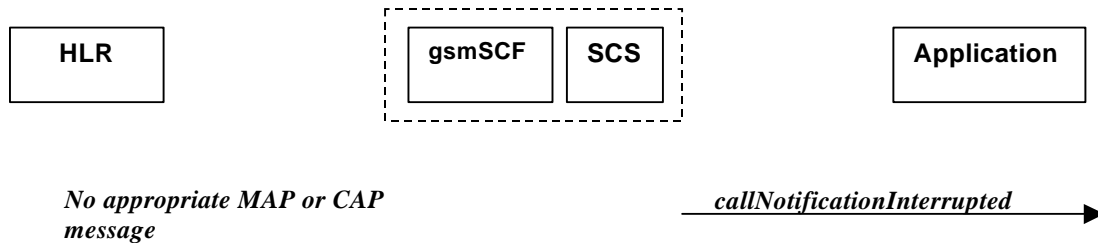
Pre-conditions	Call Control Manager is in active state
1	The application invokes the <i>setCallLoadControl</i> method
2	The SCS sends an equivalent message to the gsmSCF
3	The gsmSCF may invoke the CAP <i>CallGap</i> operations towards different gsmSSFs. CallGap can be sent in CAP only after the dialogue has been opened first by sending InitialDP.

**Table 4-10: Parameter Mapping**

From: setCallLoadControl	To: CAP CallGap
duration	gapIndicators duration
mechanism callLoadControlPerInterval	gapIndicators gapInterval
treatment ReleaseCause AdditionalTreatmentInfo InformationToSend	gapTreatment ReleaseCause  InformationToSend
addressRange	gapCriteria basicGapCriteria calledAddressValue
assignmentID	

### 4.1.6 callNotificationInterrupted

*callNotificationInterrupted* indicates to the application that all event notifications have been interrupted, for example due to faults detected.



**Figure 4-6: Call Flow for callNotificationInterrupted**

**Table 4-11: Normal Operation**

Pre-conditions	Call notifications have been enabled using the <i>enableNotification</i> method on the Call Manager interface
1	The SCS has detected, or has been informed of, a fault which prevents further events from being notified
2	The SCS invokes the <i>callNotificationInterrupted</i> method

**Parameter Mapping**

None.

### 4.1.7 callNotificationContinued

*callNotificationContinued* indicates to the application that all event notifications have been previously interrupted, have now started again.

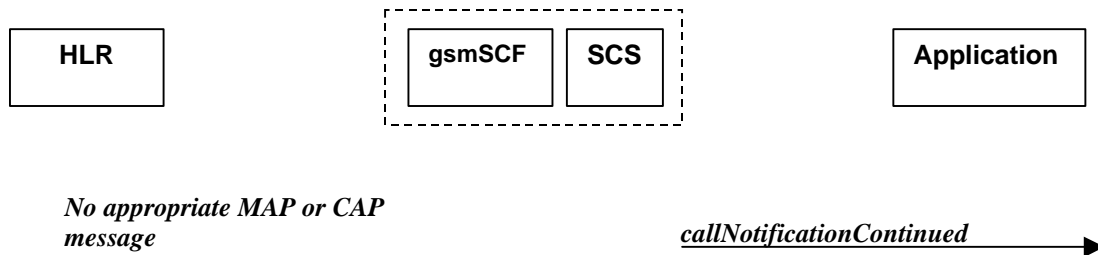


Figure 4-7: Call Flow for callNotificationContinued

Table 4-12: Normal Operation

Pre-conditions	Call notifications have been interrupted and <i>callNotificationInterrupted</i> method has been invoked.
1	The SCS detects that call notifications are again possible.
2	The SCS invokes the <i>callNotificationContinued</i> method

**Parameter Mapping**

None.

### 4.1.8 callAborted

*callAborted* indicates to the application that the call object has aborted or terminated abnormally. No further communication will be possible between the call and the application.

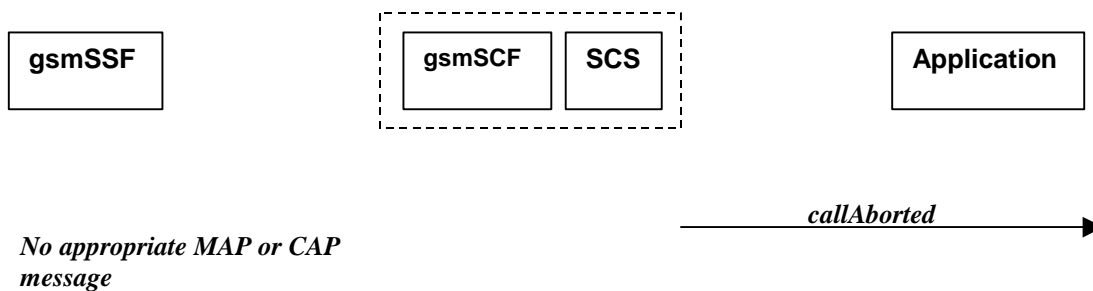


Figure 4-8: Call Flow for callAborted

Table 4-13: Normal Operation

Pre-conditions	
1	The SCS detect a catastrophic failure in its communication with the gsmSCF
2	The SCS, invokes the <i>callAborted</i> method. The call running in the network may continue and will not have been affected by this failure between the gsmSCF and the SCS

**Parameter Mapping**

None.

### 4.1.9 callEventNotify

*callEventNotify* notifies the application of the arrival of a call-related event.

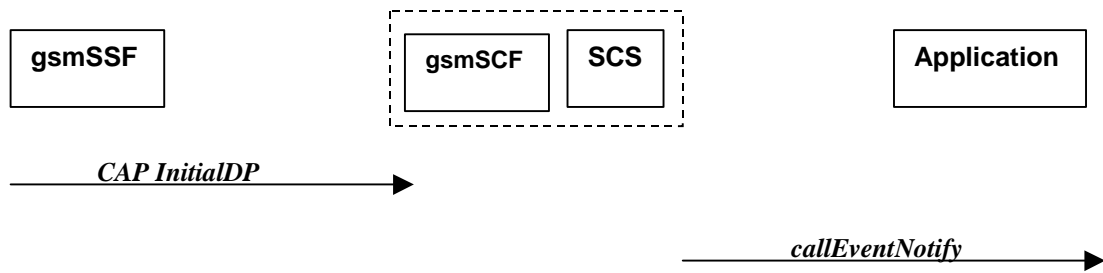


Figure 4-9: Call Flow for callEventNotify

Table 4-14: Normal Operation

Pre-conditions	Call notifications have been enabled using the <i>enableCallNotification</i> method on the Call Manager interface
1	A call arrives at the gsmSSF causing initial triggering to the gsmSCF CAP <i>InitialDP</i>
2	The gsmSCF recognizes the need for an API service and passes the triggering information to the SCS
3	The SCS identifies the application responsible for handling the call and invokes the <i>callEventNotify</i> method

Table 4-15: Parameter Mapping

From: CAP InitialDP	To: <i>callEventNotify</i>
	callReference
	eventInfo (TpCallEventInfo) :
	destinationAddress
calledPartyNumber	
calledPartyBCDNumber BCD	
calling Party Number	originatingAddress
originalCalledPartyID	originalDestinationAddress
redirectingPartyID	redirectingAddress
	callAppInfo (TpCallAppInfoSet) :
	CallAppAlertingMechanism
	CallAppNetworkAccessType
	CallAppInterworkingIndicators
ext-BasicServiceCode (1 <sup>st</sup> priority)	CallAppBearerService
	CallAppTeleService
highLayerCompatibility (2 <sup>nd</sup> priority)	CallAppTeleService
bearerCapability (2 <sup>nd</sup> priority)	CallAppBearerService
callingPartysCategory	CallAppPartyCategory
	CallAppPresentationAddress
	CallAppGenericInfo
additionalCallingPartyNumber	CallAppAdditionalAddress
eventTypeBCSM	callEventName (see Table 4-14 below)
	callNotificationType
	assignmentID
	applInterface
serviceKey	<Note: mapped to the method invocation>
cGEncountered	
iPSSPCapabilities	
locationNumber	
redirectionInformation	
iMSI	
subscriberState	
locationInformation	
callReferenceNumber	
serviceInteractionIndicatorsTwo	
mscAddress	
timeAndTimezone	
gsm-ForwardingPending	
initialDPargExtension :	
naCarrierInformation	
gmscAddress	
cause	
cug-Index	
cug-Interlock	
cug-OutgoingAccess	

Table 4-16: eventTypeBCSM mapping to callEventName

From: CAP InitialDP parameter eventTypeBCSM	To: callEventNotify parameter callEventName in eventInfo
<no mapping available>	P_EVENT_NAME_UNDEFINED
<no mapping available>	P_EVENT_GCCS_OFFHOOK_EVENT
collectedInfo, termAttemptAuthorized	P_EVENT_GCCS_ADDRESS_COLLECTED_EVENT
analyzedInformation	P_EVENT_GCCS_ADDRESS_ANALYSED_EVENT
tBusy	P_EVENT_GCCS_CALLED_PARTY_BUSY
tBusy (see Note)	P_EVENT_GCCS_CALLED_PARTY_UNREACHABLE
tNoAnswer	P_EVENT_GCCS_NO_ANSWER_FROM_CALLED_PARTY
routeSelectFailure	P_EVENT_GCCS_ROUTE_SELECT_FAILURE
<no mapping available>	P_EVENT_GCCS_ANSWER_FROM_CALL_PARTY
NOTE:	Depending on the value of the <i>cause</i> parameter in the <i>initialDPArg extensions</i> parameter of the InitialDP operation

## 4.2 Call

The generic call interface represents the interface to the generic call SCF. It provides a structure to allow simple and complex call behaviour.

### 4.2.1 routeReq

*routeReq* is an asynchronous method which requests routing of the call (and inherently attached parties) to the destination party, via a passive call leg

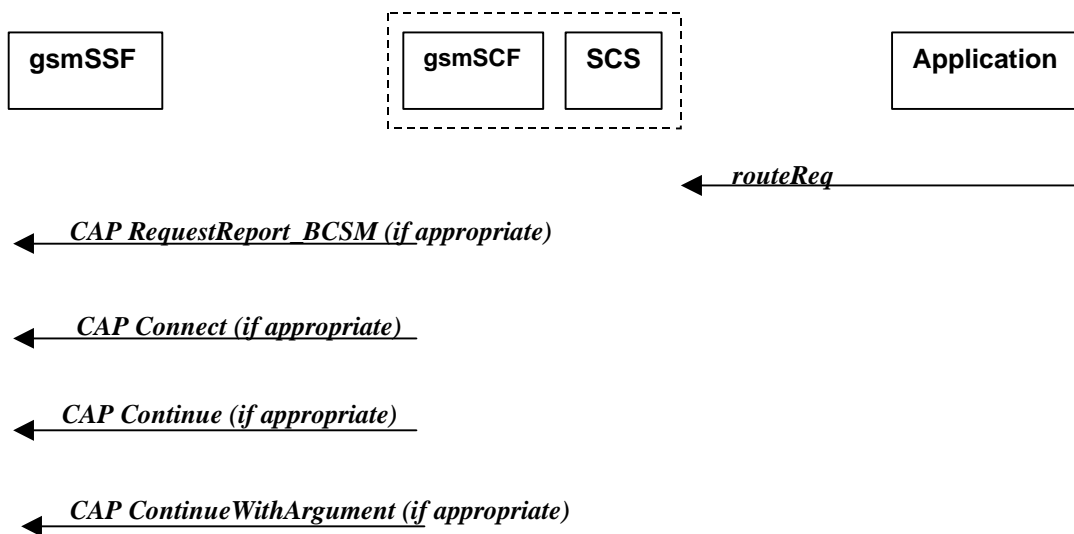


Figure 4-10: Call Flow for routeReq



**Table 4-17: Normal Operation**

Three alternatives have been identified

1. The application changes the destination number

Pre-conditions	The application has been notified of a new call and the call object exists. The <i>setCallChargePlan</i> and <i>getCallInfoReq</i> methods may have been invoked
1	The application invokes the <b>routeReq</b> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <b>RequestReportBSCM</b> if the application needs to be informed about the outcome of the request
4	The gsmSCF sends a CAP <b>Connect</b> message

**Table 4-18: Parameter Mapping**

From: routeReq	To: CAP RequestReportBCSMEvent
callSessionID	
responseRequested (TpCallReportRequestSet) :	bcsmevent :
<b>MonitorMode</b> (TpCallMonitorMode)	<b>monitorMode</b>
<b>CallReportType</b> (TpCallReportType)	<b>eventTypeBCSM</b>
<b>AdditionalReportCriteria</b> (TpCallReportAdditionalCriteria) :	dPSpecificCriteria :
<b>noAnswerDuration</b>	applicationTimer
<b>serviceCode</b>	
	<b>legID (see Note)</b>
targetAddress	
originatingAddress	
originalDestinationAddress	
redirectingAddress	
appInfo	
callLegSessionID	
NOTE: The legID for both the originating and the terminating leg are required for the disconnect event.	

Table 4-19:

From: routeReq	To: CAP Connect
callSessionID	
responseRequested	
targetAddress	destinationRoutingAddress
originatingAddress	
originalDestinationAddress	originalCalledPartyID
redirectingAddress	redirectingPartyID
appInfo (TpCallAppInfoSet) :	
CallAppAlertingMechanism	alertingPattern
CallAppNetworkAccessType	
<b>CallAppInterworkingIndicators</b>	serviceInteractionIndicatorsTwo
CallAppTeleService	
CallAppBearerService	
<b>CallAppPartyCategory</b>	callingPartysCategory
PresentationAddress	genericNumbers (see Note)
CallAppGenericInfo	
CallAppAdditionalAddress	genericNumbers
callLegSessionID	
	redirectionInformation
	suppressionOfAnnouncement
	oCSIApplicable
	na-Info :
	naCarrierInformation
	naOliInfo
	naChargeNumber
	connectArgExtension :
	cug-Interlock
	cug-OutgoingAccess
	nonCug-Call
NOTE: Operator specific function if CallAppAdditionalAddress is not used to map the genericNumbers parameter.	

Table 4-20:

2. The application does not modify the destination address and does not provide any Application Information

Pre-conditions	The application has been notified of a new call and the call object exists. The <i>setCallChargePlan</i> and <i>getCallInfoReq</i> methods may have been invoked
1	The application invokes the <b>routeReq</b> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <b>RequestReportBSCM</b> if the application needs to be informed about the outcome of the request
4	The gsmSCF sends a CAP <b>Continue</b> message

Table 4-21: Parameter Mapping

From: routeReq	To: CAP RequestReportBCSMEvent
callSessionID	
responseRequested (TpCallReportRequestSet) :	bcsmevent :
<b>MonitorMode</b> (TpCallMonitorMode)	<b>monitorMode</b>
<b>CallReportType</b> (TpCallReportType)	<b>eventTypeBCSM</b>
<b>AdditionalReportCriteria</b> (TpCallReportAdditionalCriteria) :	<b>dPSpecificCriteria</b> :
<b>noAnswerDuration</b>	applicationTimer
<b>serviceCode</b>	
	legID (see Note)
targetAddress	
originatingAddress	
originalDestinationAddress	
redirectingAddress	
applInfo	
callLegSessionID	
NOTE: The legID for both the originating and the terminating leg are required for the disconnect event.	

Table 4-22:

From: routeReq	To: CAP Continue
callSessionID	
responseRequested	
targetAddress	
originatingAddress	
originalDestinationAddress	
redirectingAddress	
applInfo	
callLegSessionID	

Table 4-23:

3. The application does not modify the destination party number but modifies Application information

Pre-conditions	The application has been notified of a new call and the call object exists. The <i>setCallChargePlan</i> and <i>getCallInfoReq</i> methods may have been invoked
1	The application invokes the <b>routeReq</b> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <b>RequestReportBSCM</b> if the application needs to be informed about the outcome of the request
4	The gsmSCF sends a CAP <b>ContinueWithArgument</b> message

Table 4-24: Parameter Mapping

From: routeReq	To: CAP RequestReportBCSMEvent
callSessionID	
responseRequested (TpCallReportRequestSet) :	bcsmEvent :
<b>MonitorMode</b> (TpCallMonitorMode)	<b>monitorMode</b>
<b>CallReportType</b> (TpCallReportType)	<b>eventTypeBCSM</b>
<b>AdditionalReportCriteria</b> (TpCallReportAdditionalCriteria) :	<b>dPSpecificCriteria</b> :
<b>noAnswerDuration</b>	applicationTimer
<b>serviceCode</b>	
	legID (see Note)
targetAddress	
originatingAddress	
originalDestinationAddress	
redirectingAddress	
appInfo	
callLegSessionID	
NOTE: The legID for both the originating and the terminating leg are required for the disconnect event.	

Table 4-25:

From: routeReq	To: CAP ContinueWithArgument
callSessionID	
responseRequested	
targetAddress	
originatingAddress	
originalDestinationAddress	
redirectingAddress	
appInfo :	
CallAppAlertingMechanism	alerting Pattern
CallAppNetworkAccessType	
CallAppInterworkingIndicators	serviceInteractionIndicatorsTwo
CallAppTeleService	
CallAppBearerService	
CallAppPartyCategory	callingPartysCategory
PresentationAddress	genericNumbers (see Note)
CallAppGenericInfo	
CallAppAdditionalAddress	genericNumbers
callLegSessionID	
	suppressionOfAnnouncement
	na-Info :
	naCarrierInformation
	naOliInfo
	naChargeNumber
	continueWithArgumentArgExtension :
	cug-Interlock
	cug-OutgoingAccess
	nonCug-Call
NOTE: Operator specific function if CallAppAdditionalAddress is not used to map the genericNumbers parameter.	

### 4.2.2 routeRes

*routeRes* is an asynchronous method which indicates that the request to route the call to the destination was successful, and indicates the response of the destination party (for example, the call was answered, not answered, refused due to busy, etc.). For every trigger that was armed in the parameter **responseRequested** of the *routeReq* a *routeRes* method may be invoked.

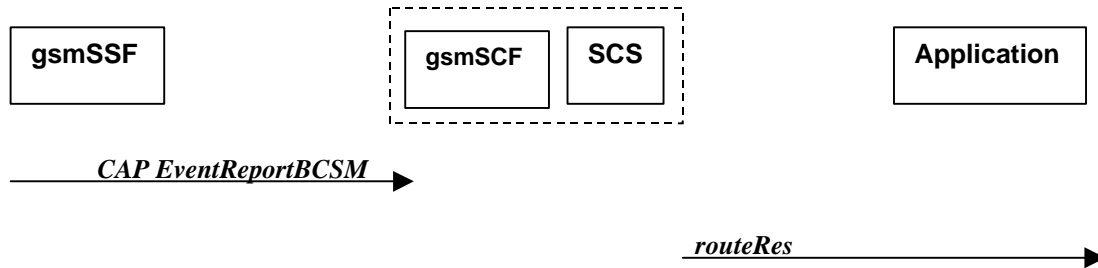


Figure 4-11: Call Flow for routeRes

Table 4-26: Normal Operation

Pre-conditions	Call routing attempted
1	If event reports have been requested, the gsmSSF sends a CAP <i>EventReportBCSM</i> to the gsmSCF
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS invokes the <i>routeRes</i> method

Table 4-27: Parameter Mapping

From: CAP EventReportBCSM	To: routeRes
	callSessionID
	eventReport :
miscCallInfo	MonitorMode
	CallEventTime
eventTypeBCSM	CallReportType (TpCallReportType)
legID	
eventSpecificInformationBCSM	AdditionalReportInfo (TpCallAdditionalReportInfo)
	callLegSessionID

### 4.2.3 routeErr

*routeErr* is an asynchronous method which indicates that the request to route the call to the destination party was unsuccessful – the call could not be routed to the destination party (for example, the network was unable to route the call, parameters were incorrect, the request was refused, etc).

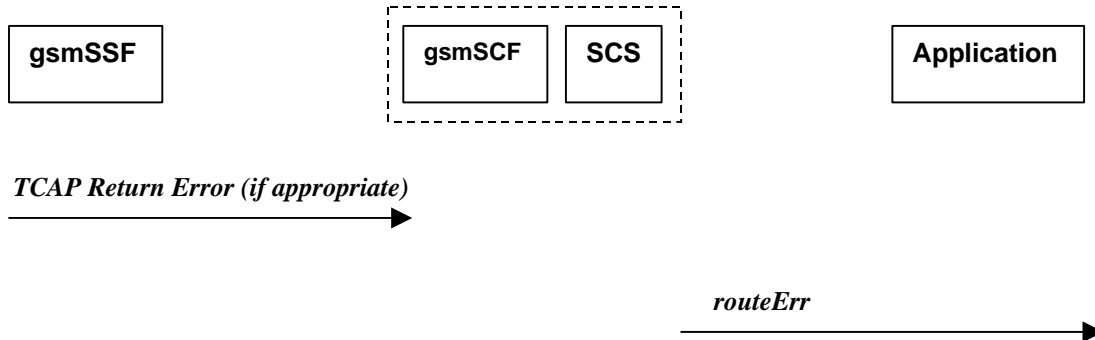


Figure 4-12: Call Flow for routeErr

Table 4-28: Normal Operation

Two scenarios are possible

1. The gsmSCF receives a message from the gsmSSF indicating an error

Pre-conditions	Call routing attempted
1	The gsmSSF detects a call routing failure and sends an appropriate TCAP message returning an error to the gsmSCF
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS detects an error with the <i>routeReq</i> method, or receives a TCAP Return Error, and invokes the <i>routeErr</i> method

Table 4-29:

2. The gsmSCF detects there is an error in the message from the SCS

Pre-conditions	Call routing attempted
1	The gsmSCF detects an error in the parameters of the internal message from the SCS requesting a <i>routeReq</i>
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS invokes the <i>routeErr</i> method

Table 4-30: Parameter Mapping

From: TCAP Return Error	To: routeErr
	callSessionID
TC-U-ERROR TC-U-REJECT	error
	callLegSessionID

### 4.2.4 release

*release* is a method used to request the release of the call and associated objects.

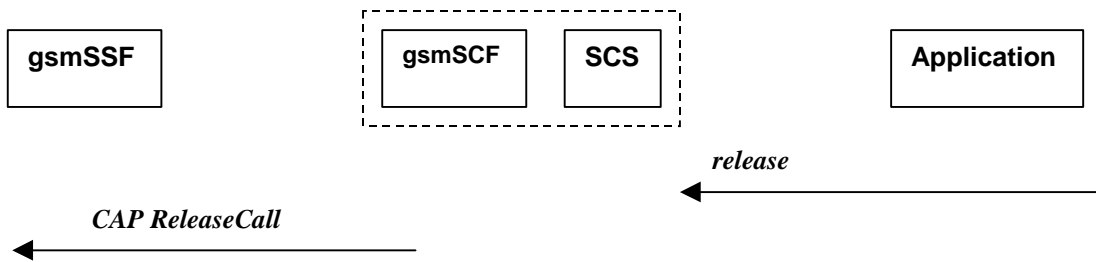


Figure 4-13: Call Flow for release

Table 4-31: Normal Operation

Pre-conditions	Call is in progress
1	The application invokes the <i>release</i> method
2	The SCS sends an equivalent message to the gsmSCF
3	The gsmSCF invokes the CAP <i>ReleaseCall</i> operation

Table 4-32: Parameter Mapping

From: release	To: CAP ReleaseCall
callSessionID	
cause (TpCallReleaseCause) :	
value (specified in ITU-T Q.850)	Cause
location	

### 4.2.5 deassignCall

*deassignCall* is a method that requests that the relationship between the application and the call and associated objects be de-assigned. It leaves the call in progress, however, it purges the specified call object so that the application has no further control of call processing. If a call is de-assigned that has event reports or call information reports requested, then these reports will be disabled and any related information discarded.

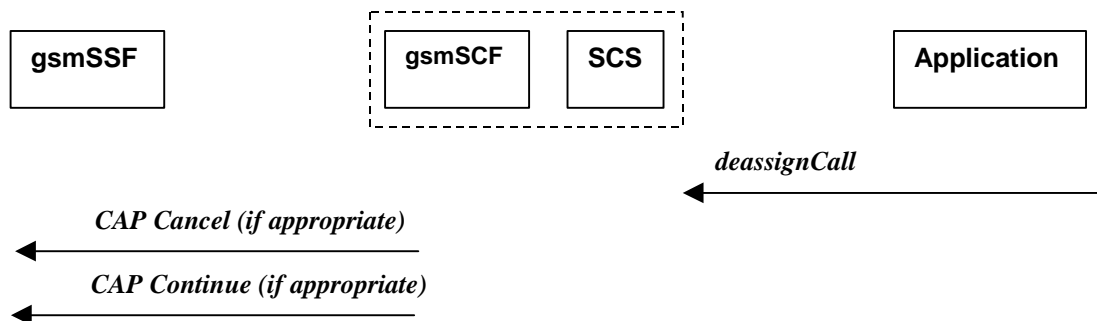


Figure 4-14: Call Flow for deassignCall

Table 4-33: Normal Operation

Pre-conditions	
1	The application invokes the <i>deassignCall</i> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <b>Cancel</b> operation to the gsmSSF if there are any reports pending.
4	The gsmSCF may send a CAP <b>Continue</b> to allow the interrupted call processing to continue. This is not sent if the call has already been established.

Table 4-34: Parameter Mapping

From: deassignCall	To: CAP Cancel
	AllRequests
callSessionID	

Table 4-35:

From: deassignCall	To: CAP Continue
callSessionID	



### 4.2.6 getCallInfoReq

*getCallInfoReq* is an asynchronous method that requests information associated with the call to be provided at the appropriate time (for example, to calculate charging). This method must be invoked before the call is routed to a target address. The call object will exist after the call is ended if information is required to be sent to the application at the end of the call. The information will be sent after any call event report.

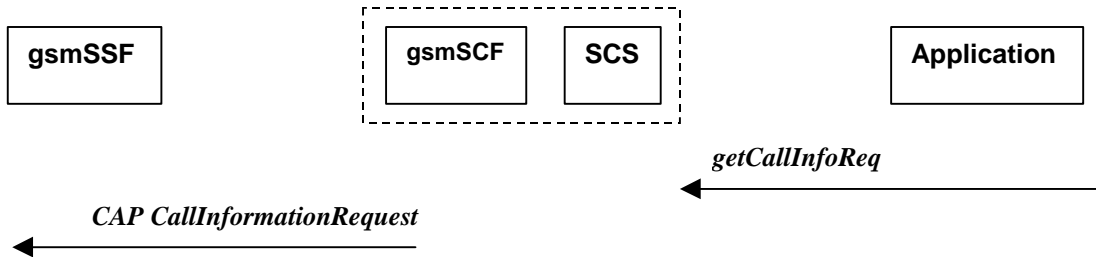


Figure 4-15: Call Flow for getCallInfoReq

Table 4-36: Normal Operation

Pre-conditions	
1	The application invokes the <i>getCallInfoReq</i> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <i>CallInformationRequest</i> operation to the gsmSSF

Table 4-37: Parameter Mapping

From: getCallInfoReq	To: CAP CallInformationRequest
callSessionID	
	RequestedInformationTypeList
<b>callInfoRequested</b> (TpCallInfoType) :	RequestedInformationType
P_CALL_INFO_UNDEFINED	
P_CALL_INFO_TIMES	callAttemptElapsedTime callStopTime callConnectedElapsedTime
P_CALL_INFO_RELEASE_CAUSE	releaseCause
P_CALL_INFO_INTERMEDIATE	
	LegID

### 4.2.7 getCallInfoRes

*getCallInfoRes* is an asynchronous method that reports all the necessary information requested by the application, for example to calculate charging.



Figure 4-16: Call Flow for getCallInfoRes

Table 4-38: Normal Operation

Pre-conditions	Call is in progress
1	The gsmSCF receives a CAP <b>CallInformationReport</b> from the gsmSSF.
2	The gsmSCF sends an equivalent internal message to the SCS
3	The SCS identifies the correct application and invokes the <b>getCallInfoRes</b> method

Table 4-39: Parameter Mapping

From: CAP CallInformationReport	To: getCallInfoRes
	callSessionID
requestedInformationList	callInfoReport :
requestedInformationType :	CallInfoType
	P_CALL_INFO_UNDEFINED
callAttemptElapsedTime callStopTime callConnectedElapsedTime	P_CALL_INFO_TIMES
releaseCause	P_CALL_INFO_RELEASE_CAUSE
	P_CALL_INFO_INTERMEDIATE
requestedInformationValue :	
	CallInitiationStartTime
callStopTimeValue	CallEndTime
	CallConnectedToResourceTime
	CallConnectedToDestinationTime
releaseCauseValue	Cause
LegID	

### 4.2.8 getCallInfoErr

*getCallInfoErr* is an asynchronous method that reports that the original request was erroneous, or resulted in an error condition.

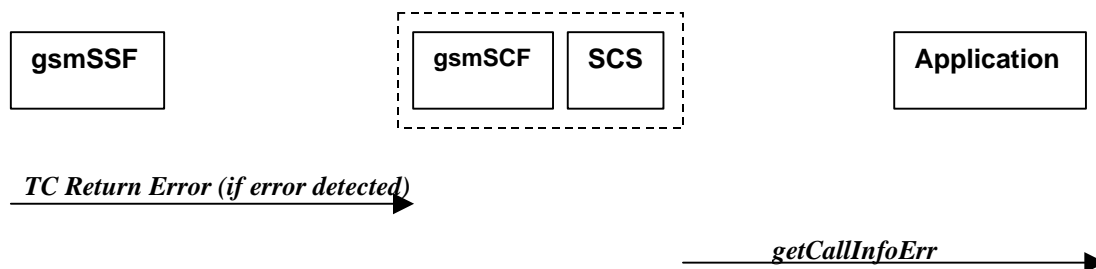


Figure 4-17: Call Flow for getCallInfoErr

Table 4-40: Normal Operation

Pre-conditions	The application has requested information associated with a call via the <i>getCallInfoReq</i> method
1	A call terminates abnormally and the gsmSSF sends an error in a TCAP message to the gsmSCF, or aborts the TCAP dialogue
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS identifies the correct applications that requested the call information and invokes the <i>getCallInfoErr</i> method.

Table 4-41: Parameter Mapping

From:	To: <i>getCallInfoErr</i>
TC Primitives	<b>callSessionID</b>
TC-U-ABORT	errorIndication
TC-P-ABORT	
TC-NOTICE	
TC-U-ERROR	
TC-L-CANCEL	
TC-U-CANCEL	
TC-L-REJECT	
TC-R-REJECT	
TC-U-REJECT	

### 4.2.9 superviseCallReq

*superviseCallReq* is a method that is called by the application to supervise a call. The application can set a granted connection time for this call. If an application calls this method before it calls a *routeReq()* or a user interaction method the time measurement will start as soon as the call is answered by the B-party or the user interaction system.

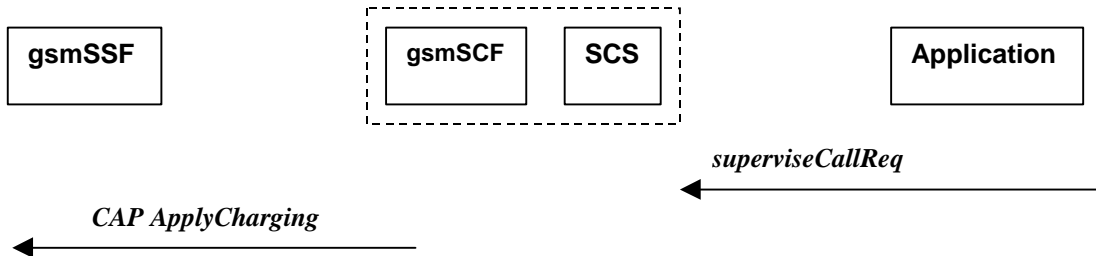


Figure 4-18: Call Flow for superviseCallReq

Table 4-42: Normal Operation

Pre-conditions	
1	The application invokes the <i>superviseCallReq</i> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <i>ApplyCharging</i> message to the gsmSSF

Table 4-43: Parameter Mapping

From: superviseCallReq	To: CAP ApplyCharging
callSessionID	
	AchBillingCharging Characteristics :
time	timeDurationCharging
	- maxCallPeriodDuration
	- tariffSwitchInterval
treatment (TpCallSuperviseTreatment) :	timeDurationCharging
P_CALL_SUPERVISE_RELEASE	- releaselfdurationExceeded
P_CALL_SUPERVISE_RESPOND	
P_CALL_SUPERVISE_APPLY_TONE	- tone
	PartyToCharge

### 4.2.10 superviseCallRes

*superviseCallRes* is an asynchronous method that reports a call supervision event to the application.

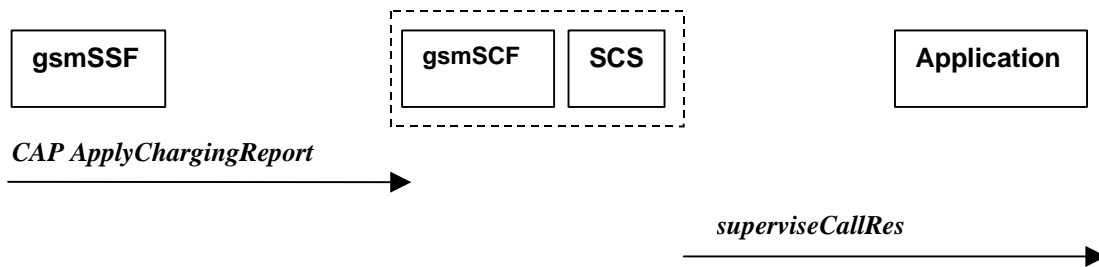


Figure 4-19: Call Flow for superviseCallRes

Table 4-44: Normal Operation

Pre-conditions	The application has invoked the supervise Call method
1	The gsmSCF receives an CAP <b>ApplyChargingReport</b> from the gsmSSF
2	The gsmSCF sends an equivalent internal message to the SCS
3	The SCS identifies the correct application and invokes the <b>superviseCallRes</b> method.

Table 4-45: Parameter Mapping

From: CAP ApplyChargingReport	To: superviseCallRes
	callSessionID
CallResult	report (TpCallSuperviseReport) :
- CallReleasedAtTcpExpiry	- P_CALL_SUPERVISE_TIMEOUT
- CallActive	- P_CALL_SUPERVISE_CALL_ENDED
	- P_CALL_SUPERVISE_TONE_APPLIED
	- P_CALL_SUPERVISE_UI_FINISHED
CallResult	usedTime
- TimeInformation	
CallResult	
- PartyToCharge	

### 4.2.11 superviseCallErr

*superviseCallErr* is an asynchronous method that reports a call supervision error to the application.

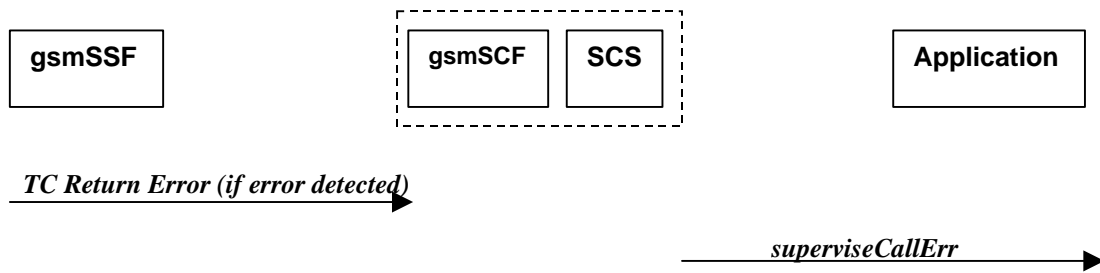


Figure 4-20: Call Flow for superviseCallErr

Table 4-46: Normal Operation

Pre-conditions	The application has requested information associated with a call via the <i>superviseCallReq</i> method
1	A call terminates abnormally and the gsmSSF sends an error in a TCAP message to the gsmSCF , or aborts the TCAP dialogue
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS identifies the correct applications that requested the call information and invokes the <i>superviseCallErr</i> method.

Table 4-47: Parameter Mapping

From:	To: <i>superviseCallErr</i>
<b>TC Primitives</b> TC-U-ABORT TC-P-ABORT TC-NOTICE TC-U-ERROR TC-L-CANCEL TC-U-CANCEL TC-L-REJECT TC-R-REJECT TC-U-REJECT	callSessionID errorIndication

### 4.2.12 setAdviceOfCharge

*setAdviceOfCharge* is a method that allows the application to determine the charging information that will be send to the end-users terminal.

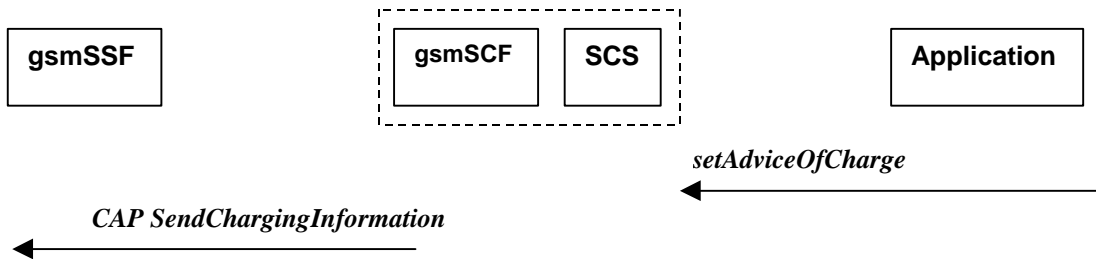


Figure 4-21: Call Flow for setAdviceOfCharge

Table 4-48: Normal Operation

Pre-conditions	
1	The application invokes the <i>setAdviceOfCharge</i> method
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <i>SendChargingInformation</i> message to the SSF

Table 4-49: Parameter Mapping

From: setAdviceOfCharge	To: CAP SendChargingInformation
callSessionID	
aOCInfo: - CurrentCAI	SCIBillingChargingCharateristics aOCBeforeAnswer aOCInitial  - or - SCIBillingChargingCharateristics aOCAfterAnswer cAI-GSM0224
- NextCAI	SCIBillingChargingCharateristics aOCBeforeAnswer aOCSubsequent cAI-GSM0224
tariffSwitch	SCIBillingChargingCharateristics aOCBeforeAnswer aOCSubsequent tariffSwitchInterval  - or - SCIBillingChargingCharateristics aOCAfterAnswer tariffSwitchInterval
	partyToCharge

### 4.2.13 setCallChargePlan

*setCallChargePlan* is a method that allows the application to include charging information in network generated CDR.

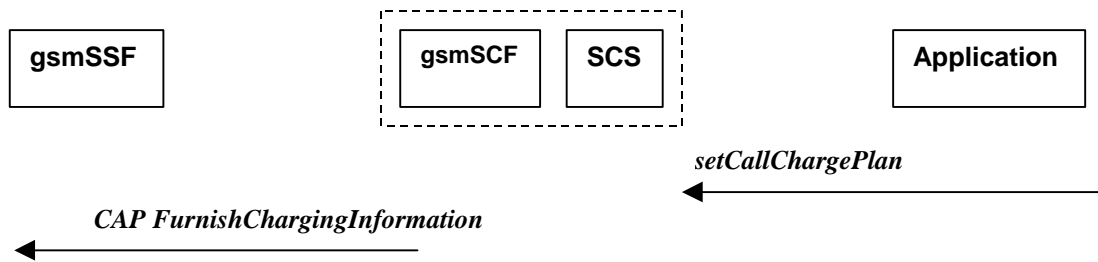


Figure 4-22: Call Flow for setCallChargePlan

Table 4-50: Normal Operation

Pre-conditions	
1	The application invokes the <i>setCallChargePlan</i>
2	The SCS sends an equivalent internal message to the gsmSCF
3	The gsmSCF sends a CAP <i>FurnishChargingInformation</i> message to the SSP

Table 4-51: Parameter Mapping

From: setCallChargePlan	To: CAP FurnishChargingInformation
callSessionID	
callChargePlan	FCIBillingChargingCharacteristics
ChargeOrderType (choice)	fCIBCCCAMELsequence1
ChargePerTime	freeFormatData
InitialCharge	
CurrentChargePerMinute	
NextChargePerMinute	
NetworkCharge	
Currency	
AdditionalInfo	
	FCIBillingChargingCharacteristics
	fCIBCCCAMELsequence1
	partyToCharge
	FCIBillingChargingCharacteristics
	fCIBCCCAMELsequence1
	appendFreeFormatData

An alternative scenario would be to map *setCallChargePlan* method to the CAP *ApplyCharging* protocol operation.



### 4.2.14 callFaultDetected

*callFaultDetected* indicates to the application that a fault has been detected in the call.

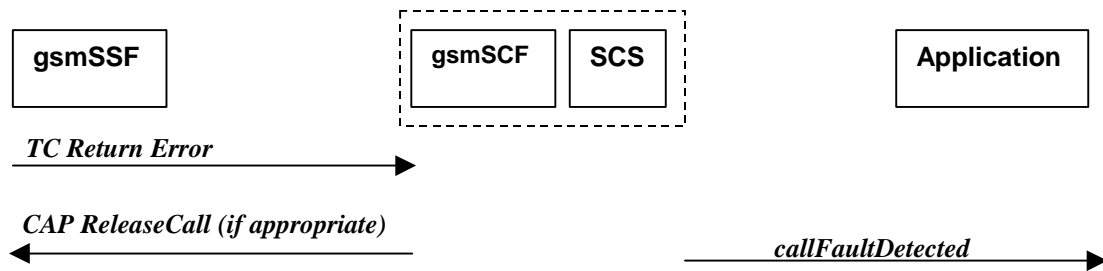


Figure 4-23: Call Flow for callFaultDetected

Table 4-52: Normal Operation

Pre-conditions	A call exists and the SCS detects an error. No routeReq method has been invoked yet.
1	The gsmSSF may detect a fault and sends an appropriate dialogue error message to the gsmSCF
2	The gsmSCF may detect a fault and send an error message to the SCS
3	The SCS detects a fault and invokes the <i>callFaultDetected</i> method
4	The SCS sends an equivalent message to the gsmSCF if appropriate
5	The gsmSCF sends a CAP <i>ReleaseCall</i> if appropriate

Table 4-53: Parameter Mapping

From: Dialogue Error	To: callFaultDetected
	callSessionID
TC_U_ABORT	fault

### 4.2.15 callEnded

*callEnded* will be invoked when the call has ended. Furthermore, the operation contains an indication on the reason why the call has been ended. Also the operation will always be invoked when the call has ended and not only when the application has requested its interest in this event.

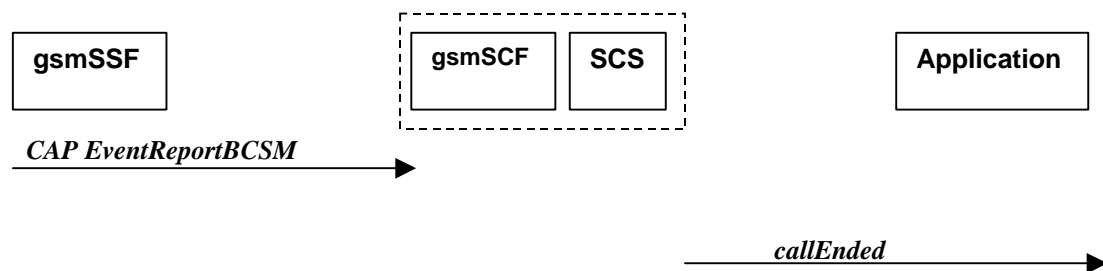


Figure 4-24: Call Flow for callEnded

Table 4-54: Normal Operation

Pre-conditions	There is an application monitoring the call in some way.
1	The gsmSSF detects a release from the calling or called party leg. CAP eventReportBCSM is sent if requested by the gsmSCF. The BCSM event indicated may be either abandon or disconnect depending on the phase of the call.
2	The gsmSCF sends an equivalent message to the SCS
3	The SCS invokes the <i>callEnded</i> method.

**Table 4-55: Parameter Mapping**

From: CAP EventReportBCSM	To: callEnded
eventTypeBCSM	callSessionID
legID	report
eventSpecificInformationBCSM: releaseCause	callLegSessionID
miscCallInfo	cause

## 5 Detailed Parameter Mappings

This clause contains detailed parameter mappings for data types that are used in the Parameter Mapping tables in the previous clauses.

### 5.1 TpCallMonitorMode

**Table 5-1:**

TpCallMonotirMode	monitorMode
P_CALL_MONITOR_MODE_INTERRUPT	interrupted
P_CALL_MONITOR_MODE_NOTIFY	notifyAndContinue
P_CALL_MONITOR_MODE_DO_NOT_MONITOR	transparent

### 5.2 TpCallReportType

**Table 5-2:**

TpCallReportType	eventTypeBCSM
P_CALL_REPORT_UNDEFINED	analyzedInformation
P_CALL_REPORT_PROGRESS	<no mapping available>
P_CALL_REPORT_ALERTING	<no mapping available>
P_CALL_REPORT_ANSWER	oAnswer tAnswer
P_CALL_REPORT_REFUSED_BUSY	oCalledPartyBusy tBusy
P_CALL_REPORT_NO_ANSWER	oNoAnswer tNoAnswer
P_CALL_REPORT_DISCONNECT	tDisconnect
P_CALL_REPORT_REDIRECTED	<no mapping available>
P_CALL_REPORT_SERVICE_CODE	<no mapping available>
P_CALL_REPORT_ROUTING_FAILURE	routeSelectFailure

## 5.3 TpCallEventName

Table 5-3:

<b>TpCallEventName</b>	<b>eventTypeBCSM</b>
P_EVENT_NAME_UNDEFINED	<no mapping available>
P_EVENT_GCCS_OFFHOOK_EVENT	<no mapping available>
P_EVENT_GCCS_ADDRESS_COLLECTED_EVENT	O-CSI (see Note) O-BcsmTriggerDetectionPoint: collectedInfo T-CSI/VT-CSI: T-BcsmTriggerDetectionPoint: termAttemptAuthorized
P_EVENT_GCCS_ADDRESS_ANALYSED_EVENT	O-CSI O-BcsmTriggerDetectionPoint analysedInfo
P_EVENT_GCCS_CALLED_PARTY_BUSY	T-CSI/VT-CSI: T-BcsmTriggerDetectionPoint: tBusy
P_EVENT_GCCS_CALLED_PARTY_UNREACHABLE	mapped to the cause value returned with TBusy : T-CSI/VT-CSI: T-BcsmTriggerDetectionPoint: tBusy
P_EVENT_GCCS_NO_ANSWER_FROM_CALLED_PARTY	T-CSI/VT-CSI: T-BcsmTriggerDetectionPoint: tNoAnswer
P_EVENT_GCCS_ROUTE_SELECT_FAILURE	O-CSI: O-BcsmTriggerDetectionPoint: routeSelectFailure
P_EVENT_GCCS_ANSWER_FROM_CALL_PARTY	T-CSI/VT-CSI: T-BcsmTriggerDetectionPoint: tAnswer
NOTE:	O-CSI applies when the value for CallNotificationType is P_ORIGINATING, T-CSI applies when the value for CallNotificationType is P_TERMINATING.

## 5.4 TpCallAdditionalReportInfo

Table 5-4:

<b>TpCallAdditionalReportInfo</b>	<b>eventSpecificInformationBCSM</b>
RefusedBusy	oCalledPartyBusy busyCause or tBusySpecificInfo busyCause callForwarded (no mapping)
CallDisconnect	oDisconnectSpecificInfo - releaseCause tDisconnectSpecificInfo - releaseCause
ForwardAddress	oAnswerSpecificInfo - destinationAddress - or-Call (no mapping) - forwardedCall (no mapping) tAnswerSpecificInfo - destinationAddress - or-Call (no mapping) - forwardedCall (no mapping)
ServiceCode	<no mapping available>
RoutingFailure	routeSelectFailureSpecificInfo - failureCause
	tNoAnswerSpecificInfo - callForwarded

---

## Annex A (informative): Change history

Change history							
Date	TSG #	TSG Doc.	CR	Rev	Subject/Comment	Old	New
Mar 2001	CN_11	NP-010131	011	--	CR 29.998: for moving TR 29.998 from R99 to Rel 4 (N5-010159)	3.2.0	4.0.0
Jun 2001	CN_12	NP-010371	001	--	Missing description of "setCallLoadControl mapping to CAP" (N5-010432)	4.0.0	4.1.0

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## History

<b>Document history</b>		
V4.0.0	March 2001	Publication
V4.1.0	June 2001	Publication