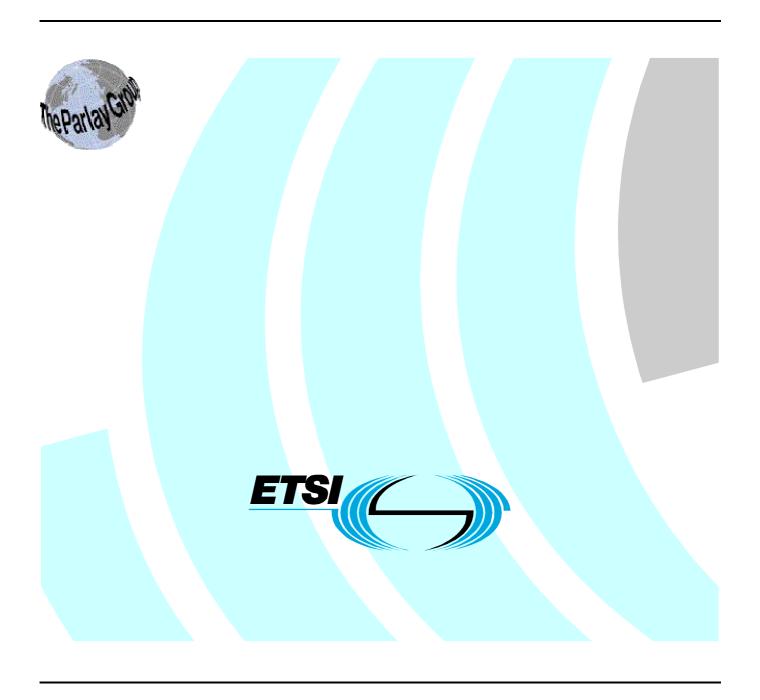
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Technical Report

Open Service Access (OSA);
Mapping of Parlay X Web Services to Parlay/OSA APIs;
Part 2: Third Party Call Mapping;
Sub-part 1: Mapping to generic control



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Foreword

NOTE:

This Technical Report (TR) has been produced by ETSI Technical Committee Telecommunications and Internet converged Services and Protocols for Advanced Networking (TISPAN).

The present document is part 2, sub-part 1, of a multi-part deliverable covering Open Service Access (OSA); Mapping of Parlay X Web Services to Parlay/OSA APIs, as identified below:

```
Part 1:
          "Common Mapping";
Part 2:
          "Third Party Call Mapping";
     Sub-part 1:
                    "Mapping to Generic Call Control";
     Sub-part 2:
                    "Mapping to Multi-Party Call Control";
Part 3:
          "Call Notification Mapping";
Part 4:
          "Short Messaging Mapping";
Part 5:
          "Multimedia Messaging Mapping";
Part 6:
          "Payment Mapping";
Part 7:
          "Account Management Mapping";
Part 8:
          "Terminal Status Mapping";
Part 9:
          "Terminal Location Mapping";
Part 10:
          "Call Handling Mapping";
Part 11:
          "Audio Call Mapping";
Part 12:
          "Multimedia Conference Mapping";
Part 14:
          "Presence Mapping";
```

the Parlay/OSA APIs. If a mapping is developed, it will become part 13 of this series.

Part 13 has not been provided as there is currently no defined mapping between ES 202 391-13 [4] and

The present document has been defined jointly between ETSI, The Parlay Group (http://www.parlay.org) and the 3GPP.

1 Scope

The Parlay X Web Services provide powerful yet simple, highly abstracted, imaginative, telecommunications functions that application developers and the IT community can both quickly comprehend and use to generate new, innovative applications.

The Open Service Access (OSA) specifications define an architecture that enables application developers to make use of network functionality through an open standardized interface, i.e. the Parlay/OSA APIs.

The present document specifies the mapping of the Parlay X Third Party Call Web Service to the Parlay/OSA Generic Call Control Service Capability Feature (SCF).

2 References

For the purposes of this Technical (TR), the following references apply:

- [1] ETSI TR 121 905: "Digital cellular telecommunications system (Phase 2+); Universal Mobile Telecommunications System (UMTS); Vocabulary for 3GPP Specifications (3GPP TR 21.905)".
- [2] W3C Recommendation (2 May 2001): "XML Schema Part 2: Datatypes".

NOTE: Available at http://www.w3.org/TR/2001/REC-xmlschema-2-20010502/.

[3] ETSI TR 102 397-1: "Open Service Access (OSA); Mapping of Parlay X Web Services to

Parlay/OSA APIs; Part 1: Common Mapping".

[4] ETSI ES 202 391-13: "Open Service Access (OSA); Parlay X Web Services; Part 13: Address List

Management".

3 Definitions and abbreviations

3.1 Definitions

For the purposes of the present document, the terms and definitions given in TR 102 397-1 [3] apply.

3.2 Abbreviations

For the purposes of the present document, the abbreviations given in TR 102 397-1 [3] apply.

4 Mapping description

The Third Party Call capability can be implemented with the Parlay/OSA Generic Call Control SCF.

It is applicable to ETSI OSA 1.x/2.x/3.x, Parlay/OSA 3.x/4.x/5.x and 3GPP Releases 4/5/6.

5 Sequence diagrams

5.1 Completed Call Released in Network

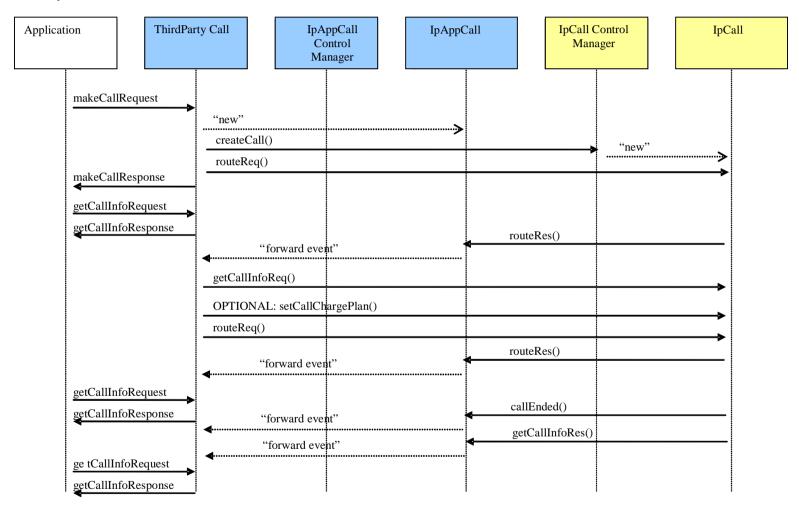


Figure 1

5.2 Completed Call Released by Application

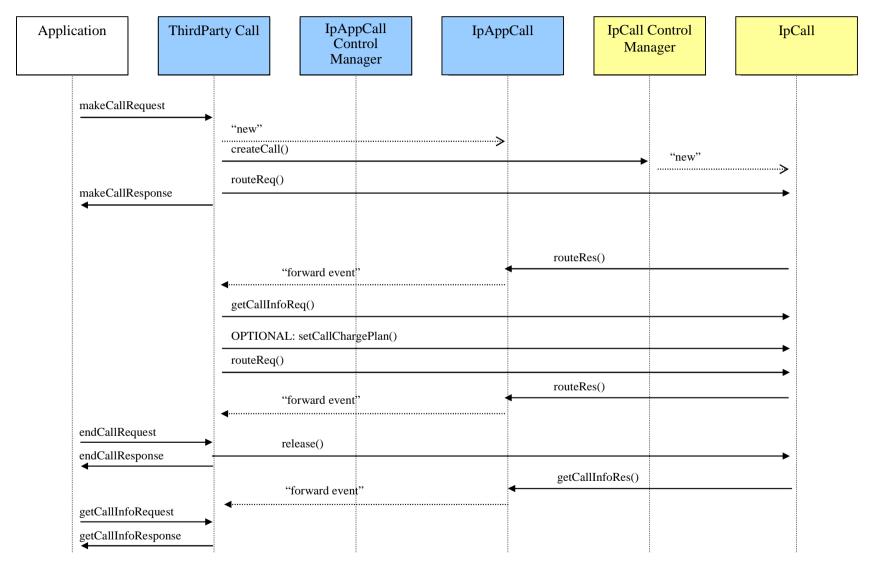


Figure 2

5.3 Call Attempt Abandoned by Application

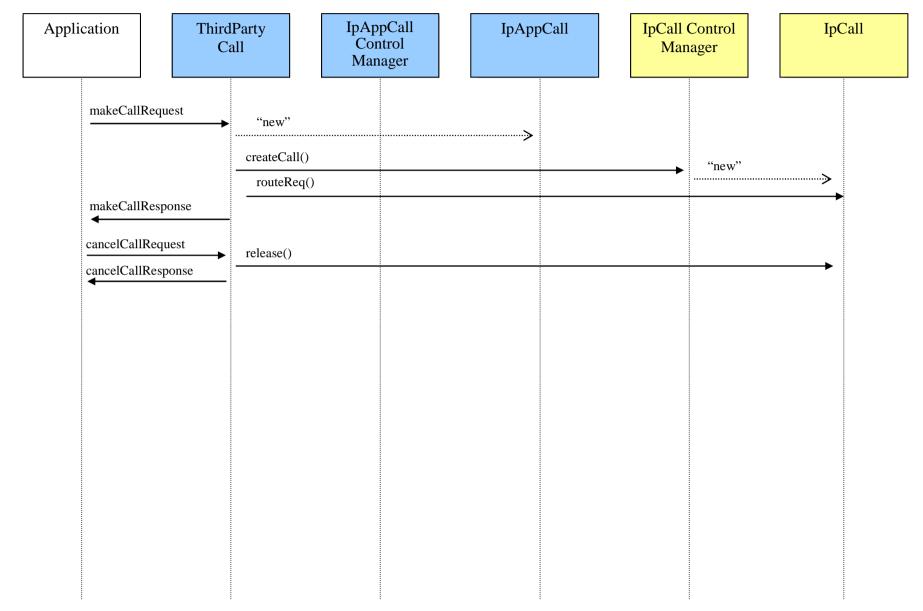


Figure 3

6 Detailed Mapping Information

6.1 Operations

6.1.1 makeCall

The sequence diagram in clause 5.1 illustrates the flow for the **makeCall** operation.

The **makeCall** operation is synchronous from the Parlay X client's point of view. It is mapped to the following Parlay/OSA methods:

- IpCallControlManager.createCall
- IpCall.routeReq
- IpCall.getCallInfoReq
- IpCall.setCallChargePlan

6.1.1.1 Mapping to IpCallControlManager.createCall

The IpCallControlManager.createCall method is invoked with the following parameters:

Name	Туре	Comment
appCall	IpAppCallRef	Reference to callback (internal)

The result from <code>IpCallControlManager.createCall</code> is of type <code>TpCallIdentifier</code> and is used internally to correlate the callbacks. Specifically it is correlated with the value of the <code>CallIdentifier</code> part returned to the application in the <code>makeCallResponse</code> message

Parlay exceptions thrown by IpCallControlManager.createCall are mapped to Parlay X exceptions as defined in clause 6.2.

6.1.1.2 Mapping to IpCall.routeReq

The IpCall.routeReq method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from lpCallControlManager.createCall]
responseRequested	TpCallReportRequest Set	Not mapped. [Requests call-related event reports: i.e. including Answer, Busy, No Answer, Not Reachable. The MonitorMode element of each requested event report should have a value of P_CALL_MONITOR_MODE_NOTIFY, with the sole exception of the P_CALL_REPORT_ANSWER event report on the calling party leg, which should have a value of P_CALL_MONITOR_MODE_INTERRUPT]
targetAddress	TpAddress	Specifies the destination leg to which the call should be routed. It is constructed based on the URI provided in the CallingParty or CalledParty part of makeCall, mapped as described in TR 102 397-1 [3]
originatingAddress	TpAddress	Specifies the calling party leg or a third party. To represent the calling party leg, the parameter value is constructed based on the URI provided in the CallingParty part of makeCall , mapped as described in TR 102 397-1 [3]. To represent a third party (e.g. a service provider number), the parameter may be populated with an address obtained from a pre-defined service level agreement.
originalDestination Address	TpAddress	Not mapped
redirectingAddress	TpAddress	Not mapped
appInfo	TpCallAppInfoSet	Not mapped

The result from <code>IpCall.routeReq</code> is of type <code>TpSessionID</code> and is used internally to correlate the callbacks. It is not mapped to the Parlay X interface.

Parlay exceptions thrown by IpCall.routeReq are mapped to Parlay X exceptions as defined in clause 6.2.

6.1.1.3 Mapping to IpCall.getCallInfoReq

The IpCall.getCallInfoReq method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from
		IpCallControlManager.createCall]
callInfoRequested	TpCallInfoType	Not mapped. [Set to a value of 03h to obtain information on
		relevant call times and release cause.]

Parlay exceptions thrown by IpCall.getCallInfoReq are mapped to Parlay X exceptions as defined in clause 6.2.

6.1.1.4 Mapping to IpCall.setCallChargePlan

The IpCall.setCallChargePlan method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from
		IpCallControlManager.createCall]
callChargePlan	TpCallChargePlan	Specifies the charge plan to use. It is constructed based on the values provided in the optional Charging part of makeCall . See the following table for details.

The callChargePlan parameter is constructed as follows:

Name	Туре	Comment
ChargeOrderType	TpCallChargeOrderCategory	Not mapped
TransparentCharge	TpOctetSet	Specifies an operator-specific charge plan. It is constructed using the value of the ChargingInformation.code element provided in the Charging part.
ChargePlan	TpInt32	Not mapped
AdditionalInfo	TpOctetSet	Descriptive string sent to billing system. It is constructed using the value of the ChargingInformation.description element provided in the Charging part. (May optionally include values of other elements of the Charging part.)
PartyToCharge	TpCallPartyToChargeType	Not mapped.
PartyToChargeAdditi onalInfo	TpCallPartyToChargeAdditional Info	Not mapped

Parlay exceptions thrown by IpCall.setCallChargePlan are mapped to Parlay X exceptions as defined in clause 6.2.

6.1.2 getCallInformation

The sequence diagram in 5.1 illustrates the flow for the **getCallInformation** operation.

The **getCallInformation** operation is synchronous from the Parlay X client's point of view. It is mapped to the following Parlay/OSA methods:

- IpAppCall.routeRes
- IpAppCall.routeErr
- IpAppCall.getCallInfoRes
- IpAppCall.getCallInfoErr
- IpAppCall.callFaultDetected
- IpAppCall.callEnded
- IpAppCallControlManager.callAborted

6.1.2.1 Mapping from IpAppCall.routeRes

The IpCall.routeRes callback method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from
		<pre>IpCallControlManager.createCall]</pre>
eventReport	TpCallReport	Specifies the result of the request to route the call to a call party, calling or called, as applicable. See the following discussion for details of its mapping to the getCallInformation operation (see note).
callLegSessionID	TpSessionID	Not mapped. [The value provide in the result from IpCall.routeReq]
NOTE: The MonitorMode and AdditionalReportInfo elements of eventReport are not mapped.		

If the **getCallInformationRequest** message is received from the application before the routeRes is invoked for either the Calling Party or Called Party, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallInitial
- StartTime, Duration, TerminationCause are not applicable

If the **getCallInformationRequest** message is received from the application after the routeRes is invoked for the Calling Party, and the call attempt was unsuccessful, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallTerminated
- StartTime, Duration are not applicable
- TerminationCause has a value mapped from the CallReportType element of the eventReport (TpCallReport) parameter as follows:
 - if the CallReportType has a value of P_CALL_REPORT_NO_ANSWER, then the **TerminationCause** has a value of **CallingPartyNoAnswer**
 - if the CallReportType has a value of P_CALL_REPORT_BUSY, then the **TerminationCause** has a value of **CallingPartyBusy**
 - if the CallReportType has a value of P_CALL_REPORT_NOT_REACHABLE, then the **TerminationCause** has a value of **CallingPartyNotReachable**
 - if the CallReportType has any other value, then the **TerminationCause** has a value of **CallAborted**

If the **getCallInformationRequest** message is received from the application after the routeRes is invoked for the Called Party, and the call attempt was unsuccessful, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallTerminated
- StartTime, Duration are not applicable
- **TerminationCause** has a value mapped from the CallReportType element of the eventReport (TpCallReport) parameter as follows:
 - if the CallReportType has a value of P_CALL_REPORT_NO_ANSWER, then the **TerminationCause** has a value of **CalledPartyNoAnswer**
 - if the CallReportType has a value of P_CALL_REPORT_BUSY, then the **TerminationCause** has a value of **CalledPartyBusy**
 - if the CallReportType has a value of P_CALL_REPORT_NOT_REACHABLE, then the **TerminationCause** has a value of **CalledPartyNotReachable**
 - if the CallReportType has any other value, then the **TerminationCause** has a value of **CallAborted**

If the <code>getCallInformationRequest</code> message is received from the application while the call between the two Parties is Active (i.e. after the <code>routeRes</code> is invoked for the Called Party, where the <code>CallReportType</code> element of the <code>eventReport</code> parameter has a value of <code>P_CALL_REPORT_ANSWER</code>), then the <code>CallInformation</code> part of the <code>getCallInformationResponse</code> message is constructed as follows:

- CallStatus has a value of CallConnected
- StartTime has the value provided in the CallEventTime element of the eventReport parameter
- **Duration** and **TerminationCause** are not applicable

6.1.2.2 Mapping from IpAppCall.routeErr

If the **getCallInformationRequest** message is received from the application before the routeErr is invoked for either the Calling Party or Called Party, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallInitial
- StartTime, Duration, TerminationCause are not applicable

If the **getCallInformationRequest** message is received from the application after the routeErr is invoked for either the Calling or Called Party, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallTerminated
- StartTime, Duration are not applicable
- TerminationCause has a value of CallAborted.

6.1.2.3 Mapping from IpAppCall.getCallInfoRes

The IpCall.getCallInfoRes callback method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from
		IpCallControlManager.createCall]
callInfoReport	TpCallInfoReport	Specifies the call information requested prior to call completion. See the following discussion for details of its mapping to the getCallInformation operation (see note).
NOTE: The CallInfoType, CallInitiationStartTime, CallConnectedToResourceTime elements of callInfoReport are not mapped.		

If the **getCallInformationRequest** message is received from the application before the IpAppCall. getCallInfoRes method is invoked, then the **CallInformation** part of the **getCallInformationResponse** message is constructed using information received from previously invoked callback methods: e.g. IpAppCall.routeRes, IpAppCall.callEnded.

If the <code>getCallInformationRequest</code> message is received from the application after invocation of the <code>IpAppCall.getCallInfoRes</code> method and within a time interval following the termination of the call between the two Parties, where this time interval is defined by the <code>StatusRetentionTime</code> service policy, then the <code>CallInformation</code> part of the <code>getCallInformationResponse</code> message is constructed as follows:

- CallStatus has a value of CallTerminated
- **StartTime** has the value provided in the CallConnectedToDestination element of the callInfoReport parameter.
- **Duration** has a value derived by subtracting the values of the CallEndTime and CallConnectedToDestination elements of the callInfoReport parameter.
- TerminationCause has a value derived from the Cause element of the callInfoReport parameter. If the Cause element has a value that explicitly indicates that one of the call parties disconnected to terminate the call, then TerminationCause has a value of CallHangUp; otherwise, TerminationCause has a value of CallAborted.

6.1.2.4 Mapping from IpAppCall.getCallInfoErr

If the **getCallInformationRequest** message is received from the application either before or after the getCallInfoErr method is invoked, then the **CallInformation** part of the **getCallInformationResponse** message is constructed using information received from previously invoked callback methods, if any: e.g. IpAppCall.routeRes/Err, IpAppCall.callEnded.

6.1.2.5 Mapping from IpAppCall.callFaultDetected

If the **getCallInformationRequest** message is received from the application after the callFaultDetected method is invoked, then the **CallInformation** part of the **getCallInformationResponse** message is constructed using information received from previously invoked callback methods: e.g. IpAppCall.routeRes/Err.

6.1.2.6 Mapping from IpAppCall.callEnded

The IpAppCall.callEnded callback method is invoked with the following parameters:

Name	Туре	Comment
callSessionID	TpSessionID	Not mapped. [The value provide in the result from
		IpCallControlManager.createCall]
Report	TpCallEndedReport	Specifies the reason the call is terminated. See the following discussion for
		details of its mapping to the getCallInformation operation (see note).
NOTE: The CallLegSessionID element of report is not mapped.		

If the **getCallInformationRequest** message is received from the application within a time interval following the termination of the call between the two Parties, where this time interval is defined by the **StatusRetentionTime** service policy, then the **CallInformation** part of the **getCallInformationResponse** message is constructed as follows:

- CallStatus has a value of CallTerminated
- StartTime. If the IpAppCall.getCallInfoRes method has already been invoked, then the value is derived from the CallConnectedToDestination element of the callInfoReport parameter of the IpAppCall.getCallInfoRes method. Otherwise the value is the date/time that the routeRes was received for the Called Party: i.e. where the CallReportType element of the eventReport parameter has a value of P_CALL_REPORT_ANSWER.
- **Duration**. If the IpAppCall.getCallInfoRes method has already been invoked, then the value is derived by subtracting the values of the CallEndTime and CallConnectedToDestination elements of the callInfoReport parameter of the IpAppCall.getCallInfoRes method. Otherwise, the value is derived by subtracting the value of **StartTime** from the date/time that the notification of call termination was received from the network, i.e. invocation of IpAppCall.callEnded.
- TerminationCause has a value derived from the Cause element of the report parameter of the IpAppCall.callEnded method. [Alternatively, if the IpAppCall.getCallInfoRes method has already been invoked, this information may be derived from the Cause element of the callInfoReport parameter of the IpAppCall.getCallInfoRes method.] In either case, if the Cause element is provided and it has a value that explicitly indicates that one of the call parties disconnected to terminate the call, then TerminationCause has a value of CallHangUp; in all other cases, TerminationCause has a value of CallAborted.

6.1.2.7 Mapping from IpAppCallControlManager.callAborted

If the **getCallInformationRequest** message is received from the application after the callAborted method is invoked, then the **CallInformation** part of the **getCallInformationResponse** message is constructed using information received from previously invoked callback methods: e.g. IpAppCall.routeRes/Err.

6.1.3 endCall

The sequence diagram in clause 5.2 illustrates the flow for the **endCall** operation.

The **endCall** operation is synchronous from the Parlay X client's point of view. It is mapped to the Parlay/OSA method: IpCall.release. If the **endCallRequest** message is received from the application after the referenced call has been terminated, it is not mapped to IpCall.release; instead a Parlay X exception is thrown: **SVC0261**.

The IpCall.release method is invoked with the following parameters:

Name	Туре	Comment	
callSessionID	TpSessionID	Not mapped. [The value provide in the result from	
		IpCallControlManager.createCall]	
cause	TpCallReleaseCause	Assigned a value indicating application-initiated call termination	

Parlay exceptions thrown by IpCall.release are mapped to Parlay X exceptions as defined in clause 6.2.

6.1.4 cancelCall

The sequence diagram in clause 5.3 illustrates the flow for the **cancelCall** operation.

The **cancelCall** operation is synchronous from the Parlay X client's point of view. It is mapped to the Parlay/OSA method: IpCall.release. If the **endCallRequest** message is received from the application after the referenced call has been connected, it is not mapped to IpCall.release; instead a Parlay X exception is thrown: **SVC0260**.

The IpCall.release method is invoked with the following parameters:

Name	Туре	Comment	
callSessionID	TpSessionID	Not mapped. [The value provide in the result from	
		<pre>IpCallControlManager.createCall]</pre>	
cause	TpCallReleaseCause Assigned a value indicating application-initiated call terminati		

Parlay exceptions thrown by IpCall.release are mapped to Parlay X exceptions as defined in clause 6.2.

6.2 Exceptions

For the present document, the mapping of Parlay/OSA API method exceptions to Parlay X Web Service exceptions is common and defined in TR 102 397-1 [3]. There are no service-specific exception mappings.

7 Additional Notes

No additional notes are provided.

History

Document history			
V1.1.1	August 2005	Publication	