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### **PacketCable™ Applications**

#### **DECT SIP Specification**

**PKT-SP-DECT-SIP-I01-090226**

**ISSUED**

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## Document Status Sheet

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## 1 SCOPE

### 1.1 Introduction and Purpose

**Note:** All references to PacketCable in this document are PacketCable 2.0 unless specified otherwise.

This document specifies the requirements for implementing HD Voice services on the PacketCable infrastructure. It focuses on the functional specification of PacketCable client devices that extend the standard telephony functions to include:

- a) DECT-based cordless telephony interface;
- b) Support for wideband voice; and
- c) Interactions between the DECT cordless handsets and analog phones.

This specification introduces two types of the HD Voice client devices. One is an embedded device that integrates a DECT base station, a PacketCable client, and a DOCSIS cable modem; the other is a stand-alone device that integrates the first two elements but not a cable modem, and is connected to the PacketCable network via an external cable modem or router. The embedded device type is similar to an E-DVA, with the exception that it supports a DECT telephony interface in addition to analog telephony interfaces.

A substantial portion of the document is devoted to the requirements on the mapping between the signaling protocol on the DECT air interface and the PacketCable SIP signaling protocol on the network side. Special considerations are given to ensure that the user experience with the existing PacketCable feature set be preserved with the HD Voice client devices.

One general goal underpinning the development of this specification is to maximize the reuse of the existing PacketCable and DECT specifications and to minimize the changes to these specifications. In particular, no change is made to the PacketCable signaling protocol on the network side in order to support the HD Voice services, thus allowing the PacketCable infrastructure to be used as is. In addition, no major change is introduced to the DECT air interface signaling protocol. Rather, the document is mainly concerned with how to employ the existing DECT signaling scheme to realize the existing telephony features and how to bridge the signaling protocols on the air interface and the PacketCable network.

The scope of this specification is limited to the requirements that impact the device's interoperability and the user's basic feature interactions. No attempt is made to specify the device's internal hardware and software architectures, the actual user interface, and supplementary functionality such as answer machine, phone book, etc. These areas, though important for a final product, are construed as either vendor-specific or operator-specific.

## 1.2 Requirements

Throughout this document, the words that are used to define the significance of particular requirements are capitalized. These words are:

"MUST"	This word means that the item is an absolute requirement of this specification.
"MUST NOT"	This phrase means that the item is an absolute prohibition of this specification.
"SHOULD"	This word means that there may exist valid reasons in particular circumstances to ignore this item, but the full implications should be understood and the case carefully weighed before choosing a different course.
"SHOULD NOT"	This phrase means that there may exist valid reasons in particular circumstances when the listed behavior is acceptable or even useful, but the full implications should be understood and the case carefully weighed before implementing any behavior described with this label.
"MAY"	This word means that this item is truly optional. One vendor may choose to include the item because a particular marketplace requires it or because it enhances the product, for example; another vendor may omit the same item.

## 2 REFERENCES

### 2.1 Normative References

In order to claim compliance with this specification, it is necessary to conform to the following standards and other works as indicated, in addition to the other requirements of this specification. Notwithstanding, intellectual property rights may be required to use or implement such normative references.

- [CODEC MEDIA] PacketCable Codec and Media Specification, PKT-SP-CODEC-MEDIA-I05-080710, July 10, 2008, Cable Television Laboratories, Inc.
- [DECT-HDV] PacketCable High Definition Voice with DECT Specification, PKT-SP-DECT-HDV-I01090226I01-090226, February 26, 2009, Cable Television Laboratories, Inc.
- [DECT-PROV] PacketCable DECT Provisioning Specification, PKT-SP-DECT-PROV-I01090226I01-090226, February 26, 2009, Cable Television Laboratories, Inc.
- [E-DVA] PacketCable Residential SIP Telephony E-DVA Specification, PKT-SP-RST-E-DVA-I04-080710, July 10, 2008, Cable Television Laboratories, Inc.
- [EN 300 444] ETSI EN 300 444, Digital Enhanced Cordless Telecommunications (DECT); Generic Access Profile (GAP), V2.1.1 (2008-10).
- [PKT 24.229] PacketCable SIP and SDP Stage 3 Specification 3GPP TS 24.229, PKT-SP-24.229-I04-080425, April 25, 2008, Cable Television Laboratories, Inc.
- [RSTF] PacketCable Residential SIP Telephony Feature Specification, PKT-SP-RSTF-I04-080710, July 10, 2008, Cable Television Laboratories, Inc.

### 2.2 Informative References

This specification uses the following informative references.

- [ARCH-FRM TR] PacketCable Architecture Framework Technical Report, PKT-TR-ARCH-FRM-V05-080425, April 25, 2008, Cable Television Laboratories, Inc.
- [CMCI] DOCSIS Cable Modem to Custom Premise Equipment Interface, CM-SP-CMCI-C01-081104, November 4, 2008, Cable Television Laboratories, Inc.
- [eDOCSIS] eDOCSIS Specification, CM-SP-eDOCSIS-I17-090121, January 21, 2009, Cable Television Laboratories, Inc.

### 2.3 Reference Acquisition

- Cable Television Laboratories, Inc., 858 Coal Creek Circle, Louisville, CO 80027; Phone +1-303-661-9100; Fax +1-303-661-9199; <http://www.cablelabs.com>
- European Telecommunications Standards Institute (ETSI), 650, route des Lucioles, 06921 Sophia-Antipolis Cedex, France, Tel.: +33 (0) 4 92 94 42 00, Fax: +33 (0) 4 93 65 47 16, <http://www.etsi.org/WebSite/homepage.aspx>.

### 3 TERMS AND DEFINITIONS

This specification uses the following terms:

<b>Digital Voice Adaptor</b>	Logical entity that is used to represent either E-DVA or S-DVA
<b>Embedded Digital Voice Adaptor</b>	The eDOCSIS device that integrates a DECT base station (FP) with a PacketCable E-DVA (embedded with cable modem)
<b>Fixed Part</b>	The DECT base station
<b>High Definition Voice</b>	The PacketCable service that offers wideband voice using DECT-based custom premise equipment
<b>Information Element</b>	The constituent part of a DECT message
<b>PacketCable Network</b>	The network that provides PacketCable services according to [ARCH-FRM TR]
<b>Portable Part</b>	The DECT handset
<b>Standalone Digital Voice Adaptor</b>	The stand-alone device that integrates a DECT base station (FP) with a PacketCable client, separate from cable modem

## 4 ABBREVIATIONS AND ACRONYMS

This specification uses the following abbreviations:

<b>AS</b>	Application Server
<b>DECT</b>	Digital Enhanced Cordless Telecommunications
<b>DOCSIS</b>	Data-Over-Cable Service Interface Specification
<b>DVA</b>	Digital Voice Adaptor
<b>E-DVA</b>	Embedded Digital Voice Adaptor
<b>eDOCSIS</b>	Embedded DOCSIS
<b>FP</b>	Fixed Part
<b>HD</b>	High Definition
<b>IE</b>	Information Element
<b>POTS</b>	Plain Old Telephone Service
<b>PP</b>	Portable Part
<b>RST</b>	Residential SIP Telephony
<b>S-DVA</b>	Standalone Digital Voice Adaptor
<b>SIP</b>	Session Initiation Protocol

## 5 OVERVIEW

This specification introduces two PacketCable client devices that allow the HD Voice services to be supported over the PacketCable network. In this document, the terms E-DVA and S-DVA are used as follows.

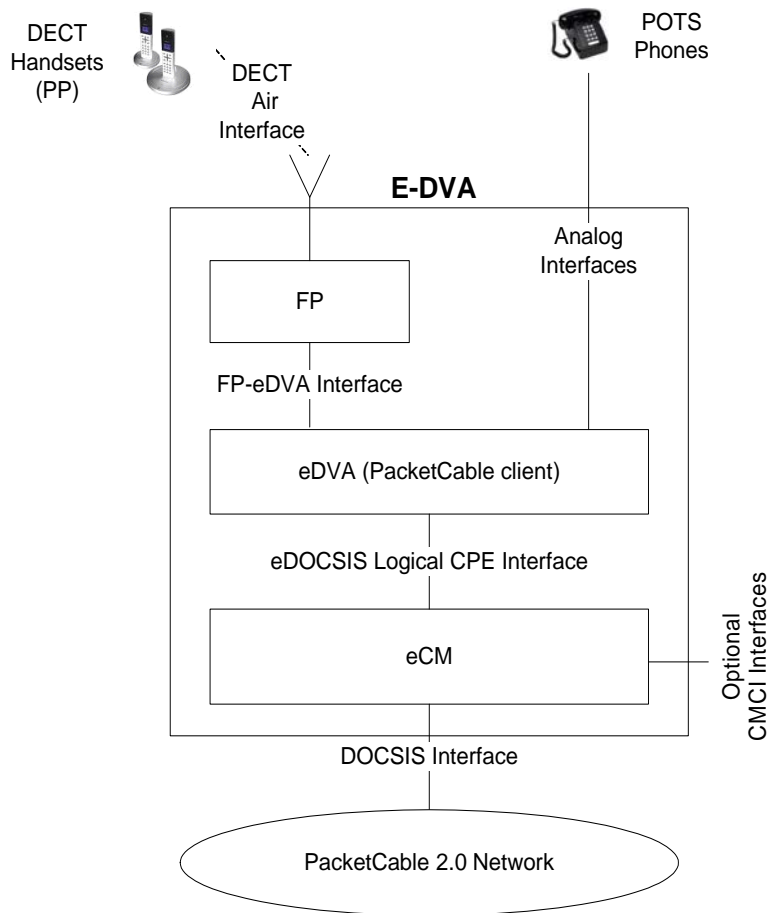
**E-DVA:** This is an eDOCSIS device with embedded cable modem, and is similar to a PacketCable E-DVA, with the main difference that it has a DECT base station embedded to provide the cordless telephony interface.

**S-DVA:** This is a standalone device with DECT base station and PacketCable client. It connects to the PacketCable network via an external cable modem.

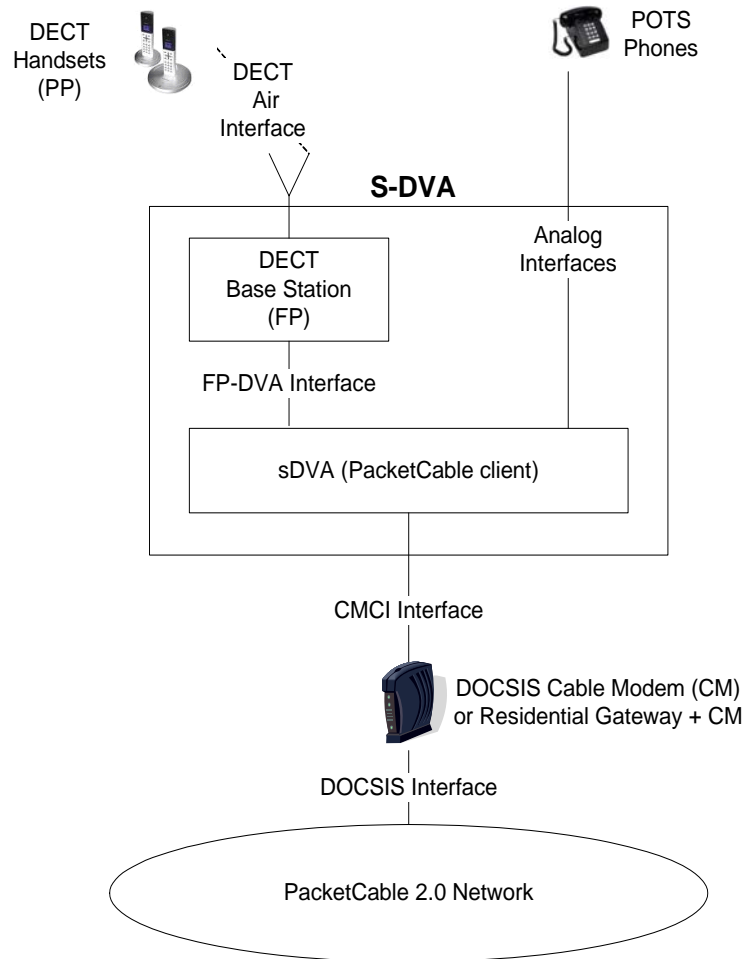
The requirements in this document apply to both device types. To simplify the presentation, the term DVA is used to represent both variants of the device, unless explicitly stated otherwise.

### 5.1 DECT-SIP Reference Architecture

The high-level functional block diagrams of E-DVA and S-DVA are shown in Figure 1 and Figure 2.



**Figure 1 - E-DVA Reference Block Diagram**



**Figure 2 - S-DVA Reference Block Diagram**

## 5.1.1 Functional Components

### 5.1.1.1 PP

The DECT handset that is compliant with this specification. It provides the necessary user interface to the HD Voice services.

### 5.1.1.2 FP

The DECT base station embedded in an E-DVA or S-DVA. It supports one or more PPs, and interacts with eDVA or sDVA in mapping between the DECT and PacketCable signaling protocols.

### 5.1.1.3 eDVA

The eSAFE in an E-DVA that represents the PacketCable client to the network [eDOCSIS]. It interacts with the FP in mapping between the DECT and PacketCable signaling protocols.

### 5.1.1.4 sDVA

The logical component inside an S-DVA that represents the PacketCable client to the network. It interacts with the FP in mapping between the DECT and PacketCable signaling protocols.

### **5.1.1.5 eCM**

The embedded DOCSIS cable modem [eDOCSIS]. To support the HD Voice services, eCM needs to be compliant with DOCSIS 1.1 or higher.

### **5.1.1.6 Cable Modem**

The stand-alone cable modem that is used to connect the S-DVA to the PacketCable Network.

### **5.1.1.7 Residential Gateway**

The gateway device that may be used to connect the S-DVA to the PacketCable Network via the cable modem.

### **5.1.1.8 Analog Phone**

The POTS phone that provides the traditional telephony user interface.

### **5.1.1.9 PacketCable Network**

The standard PacketCable compliant SIP network [ARCH-FRM TR]. No special requirements or restrictions are imposed on this network by HD Voice.

## **5.1.2 Protocol Interfaces**

### **5.1.2.1 eDOCSIS Logical CPE Interface**

The standard logical interface between the eSAFE and the eCM [eDOCSIS]. This interface is mandatory for E-DVA.

### **5.1.2.2 DOCSIS Interface**

The standard broadband interface that is compliant with DOCSIS 1.1 or higher. This interface is mandatory for both E-DVA and S-DVA.

### **5.1.2.3 FP-eDVA Interface**

The logical interface between the FP and the eDVA. Depending on the vendor implementation, this interface may or may not correspond to any well-defined physical entity. It simply represents the interactions between the FP and the eDVA, as specified in this document. This interface is vendor-specific.

### **5.1.2.4 Analog Interface**

The standard POTS interface that is compliant with [E-DVA]. This interface is mandatory for both E-DVA and S-DVA.

### **5.1.2.5 CMCI Interface**

The standard CPE interface that is compliant with [CMCI]. This interface is optional for both E-DVA and S-DVA.

## **5.2 Relationship to other specifications**

- PacketCable DECT specification [DECT-HDV]:

The PacketCable DECT specification mandates the relevant parts of the DECT specification for the air interface between the PP and the FP. It also defines requirements for interactions between the FP and the PacketCable client that are embedded in the client device that interfaces to the PacketCable network. All requirements in the PacketCable DECT specification are mandated for both the E-DVA and the S-DVA.

- PacketCable 24.229 delta specification [PKT 24.229]:

The PacketCable 24.229 delta specification defines SIP and SDP session requirements for the PacketCable UE and network components. This specification is application-agnostic and also includes requirements in the area of security, authentication, NAT/firewall traversal and others. This is also known as the PacketCable core network specification.

- PacketCable Residential SIP Telephony (RST) Feature specification [RSTF]:

The PacketCable Residential SIP Telephony (RST) Feature specification specifies interoperability requirements on the UE and Application Server to implement residential telephony features. All these features need to be supported on the DECT air interface. This specification leverages [PKT 24.229] for the application-independent requirements.

- PacketCable Residential SIP Telephony Embedded Digital Voice Adaptor [E-DVA]:

The PacketCable Residential SIP Telephony Embedded Digital Voice Adaptor specification specifies requirements for a device that integrates the cable modem with a PacketCable Residential SIP Telephony client that supports an analog interface to a POTS phone. It refers to [RSTF] for the feature signaling requirements and defines additional requirements in the areas of codec, security, provisioning, analog interface requirements, and others.

The current specification defines the mapping requirements between DECT and SIP for features specified in the above-mentioned documents. The current specification also defines device requirements for the E-DVA and the S-DVA. These requirements are based on [E-DVA] and enhanced to add requirements for HD voice codecs on the network interface, DECT interface provisioning, NAT traversal, and others.

## 6 DECT-SIP REQUIREMENTS

This section specifies requirements for DECT to SIP interworking. These requirements apply to the DVA as a whole, and their actual implementation involve the interactions between the FP and the eDVA (for E-DVA) or the sDVA (for S-DVA) shown in Figure 1 and Figure 2.

The DECT-SIP mapping in this section is mainly concerned with the signaling functionality. The DECT-SIP mapping in terms of voice media is straightforward and is considered vendor-specific: it involves the mapping between the voice payloads in the RTP packets on the network side and the DECT voice channel data over the DECT air interface.

### 6.1 Basic Calls

#### 6.1.1 Inbound Call

Upon receiving an INVITE for a new call when no other external call is present, the DVA MUST begin call establishment procedures as per [DECT-HDV]. The DVA populates the <<CALLING PARTY NUMBER>> and <<CALLING PARTY NAME>> IEs in the CC-SETUP or IWU-INFO based on the contents of the P-Asserted-Identity header in the received INVITE, as described in Section 6.3.2. The DVA MUST populate the <<CODEC LIST>> IE in the CC-SETUP and subsequently carry out the codec negotiation with the calling party, according to the codec negotiation requirements in [DECT-HDV].

After the ringing is applied to the analog interface, the DVA MUST send a 180 Ringing message in response to the inbound INVITE.

Upon receiving the CC-CONNECT message from one of the PPs, the DVA MUST send a 200 OK message in response to the inbound INVITE unless the call has already been answered (by another PP or analog interface).

An example call flow for an inbound call is shown in Figure 3.

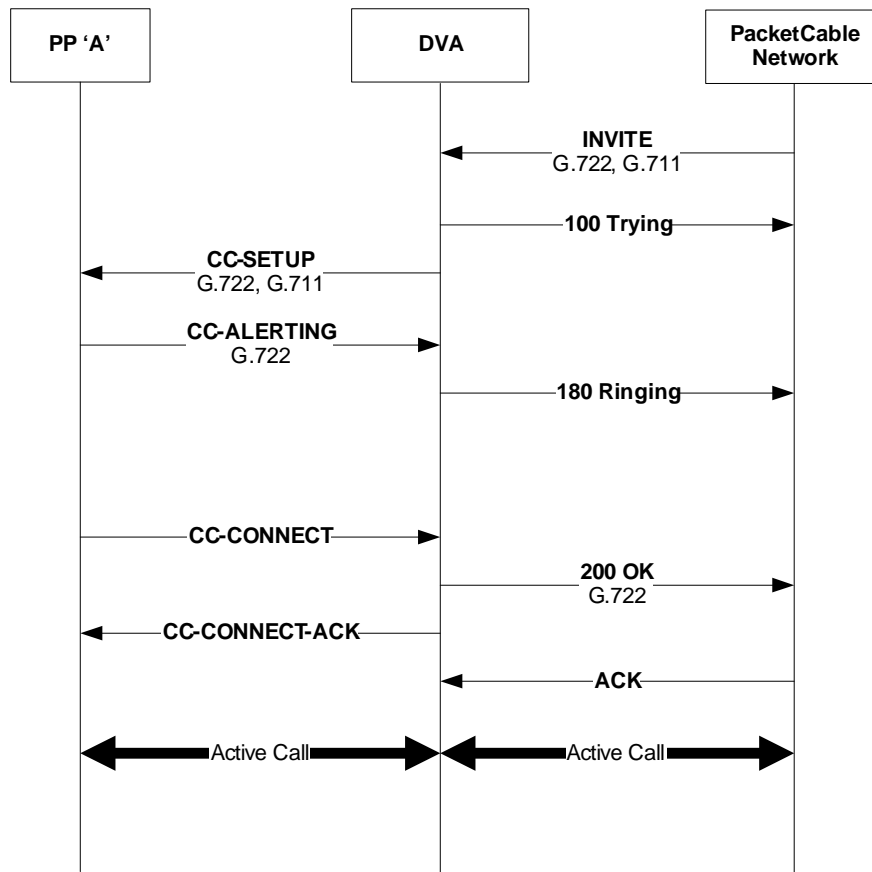


Figure 3 - Inbound call

### 6.1.2 Outbound Call

Upon receiving the CC-SETUP message without a <<CALLED PARTY NUMBER>> IE that results in DVA transition to off-hook status as per [DECT-HDV], the DVA MUST follow the procedures in [DECT-HDV] for outgoing call. This involves opening of the U plane with the CC-SETUP-ACK and presenting a dial tone to the user. This is referred to as overlap sending in [EN 300 444] and [DECT-HDV].

The PP sends dialed digits within <<KEYPAD>> or <<MULTI-KEYPAD>> IEs in one or more CC-INFO messages as specified in [DECT-HDV]. When the digits are being received, the DVA MUST apply the digit map to the received dial string as defined in Section 6.2.1 and follow the defined digit map action. Normally, the action is to send an INVITE towards the network. In this case, the DVA MUST format the INVITE as per [RSTF].

Upon the receipt of 180 Ringing from the network, the DVA MUST send a CC-ALERTING message to the PP. The DVA MUST forward a ring back tone to the PP as in-band media.

Upon the receipt of 200 OK from the network, the DVA MUST send a CC-CONNECT message to the PP.

If the remote end point is busy and cannot accept another call, the DVA receives a 486 Busy Here response to the outbound INVITE. In this case, the DVA MUST play a busy tone to the PP as in-band media.

An example call flow for an outbound call with overlap sending is shown in Figure 4.

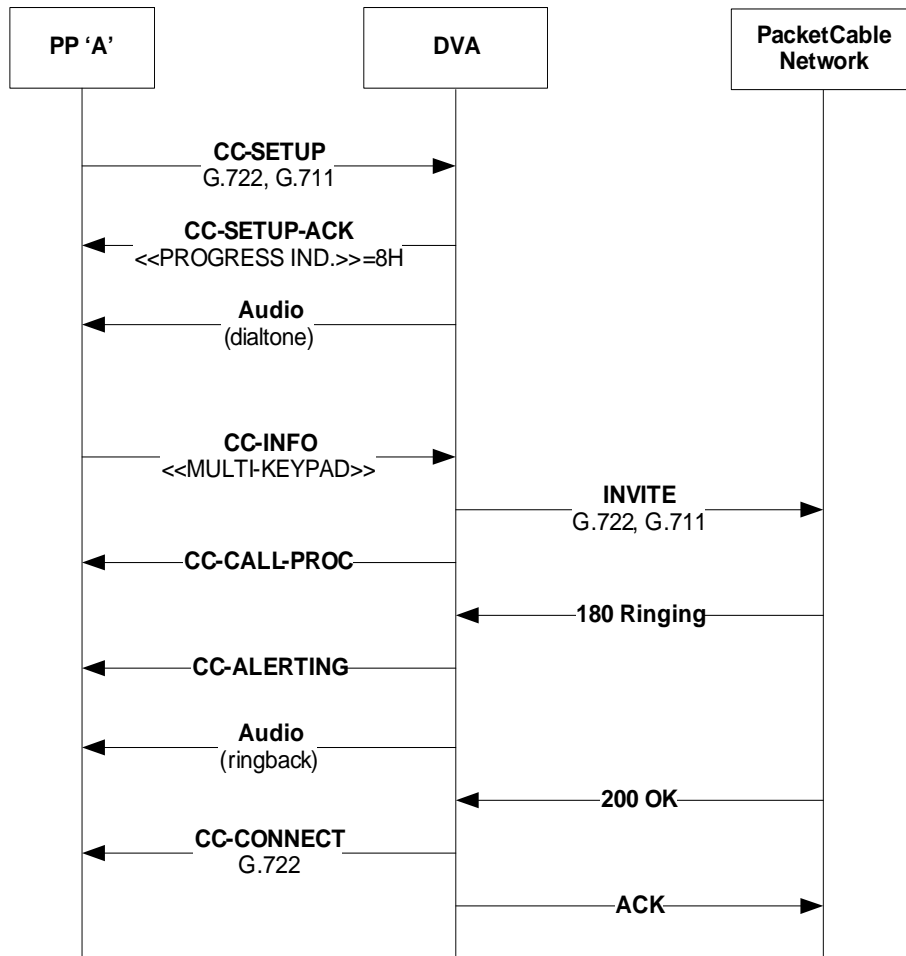


Figure 4 - Outbound call with overlap sending

### 6.1.3 PP initiated Call Release

Upon receipt of a CC-RELEASE from the PP that results in DVA transition to on-hook state as specified in [DECT-HDV], the DVA MUST send a BYE towards to the network as specified in [RSTF].

The DVA MUST respond to the CC-RELEASE with a CC-RELEASE-COM to the PP as specified in [DECT-HDV].

An example call flow for a PP initiated call release is shown in Figure 5.

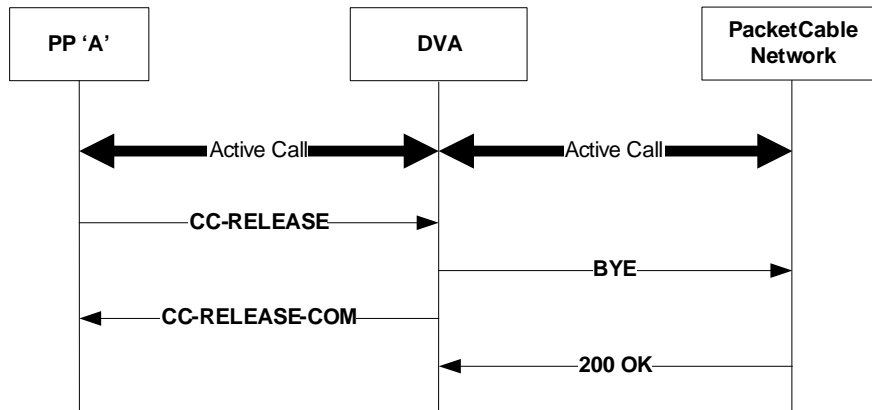


Figure 5 - PP initiated call release

### 6.1.4 Remote endpoint initiated Call Release

Upon receipt of a BYE message from the network for an active external call, the DVA MUST initiate call release procedures by sending a CC-RELEASE message to each of the PPs currently involved in that call as specified in [DECT-HDV].

An example call flow for a remote endpoint initiated call release is shown in Figure 6.

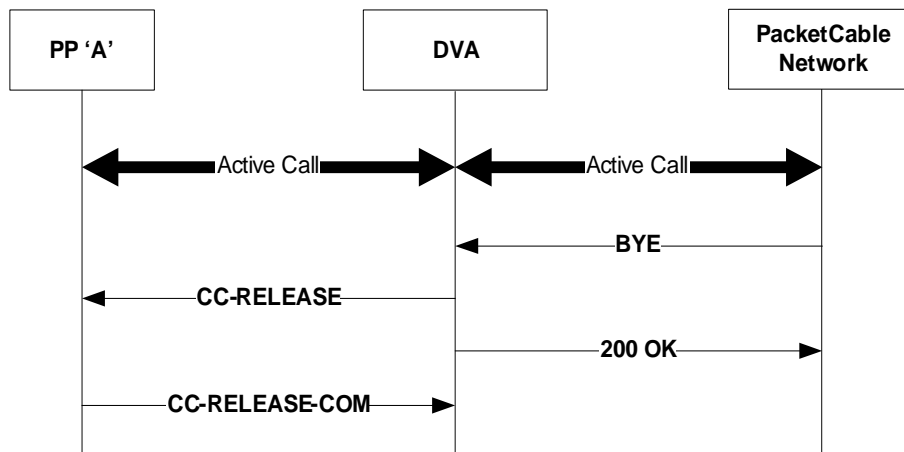


Figure 6 - Remote endpoint initiated call release

### 6.1.5 Barge into Existing External Call

A PP is automatically bridged into an existing external call between another PP and the remote PacketCable endpoint if the DVA has been configured for automatic barge-in as specified in [DECT-HDV]. The barge-in does not result in any signaling towards the PacketCable network and is handled locally on the DVA.

An example call flow for a barge-in scenario is shown in Figure 7. In this example, the PP 'A' is already involved in an external call. The PP 'B' goes off-hook, resulting in automatic barge-in. The DVA performs the media mixing function.

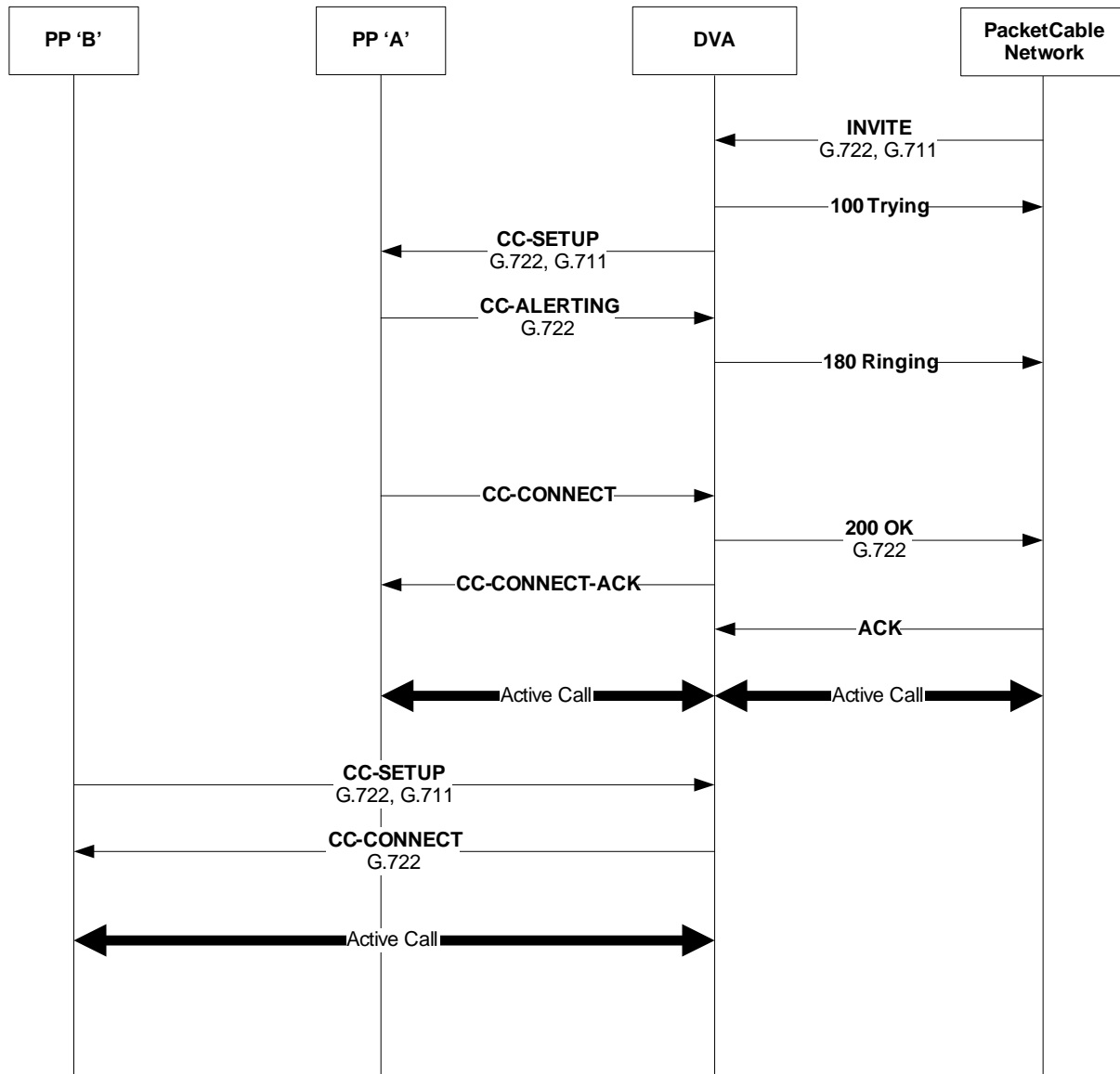


Figure 7 - Barge into existing external call

## 6.2 Common Feature Functions

The PacketCable telephony features are based on a few shared common functions. The DECT interactions for these functions are specified below.

### 6.2.1 Digit Maps

Digit maps are provisioned on the DVA to inform it about how digits entered on the PP by the user should be interpreted. A digit map is an ordered set of regular expressions combined with some special tokens that represent actions to be carried out by the DVA when a regular expression is matched.

The PP forwards the user-entered digits to the DVA by following the corresponding procedure defined in [DECT-HDV]. As the digits are being received from the PP, the DVA MUST apply the provisioned digit map to the digits according to the syntax and semantics defined in [RSTF].

### 6.2.2 Vertical Service Codes

Vertical service codes (VSC) are dialed digit strings used to represent feature invocations. They are sometimes referred to as star codes or feature activation codes. A feature invocation is feature activation, feature deactivation, feature programming, or feature instance. A VSC does not have to begin with a star, nor is it restricted to a 3-digit string such as "\*69." The actual digits that make up a VSC are defined by the digit map. The digit map also defines whether the VSC is processed by the DVA or forwarded to the network.

The PP forwards a VSC as user-entered digits to the FP in the DVA by following the corresponding procedure defined in [EN 300 444]. As the VSC digits are being received from the PP, the DVA applies the provisioned digit map to the digits as specified in Section 6.2.1.

### 6.2.3 Held Media

There are a number of service capabilities that make use of a call state that is 'held' and, as a result, also require that any active bi-directional media stream also be held. [RSTF] specifies media hold and retrieval in terms of SDP offer-answer procedures.

When service logic requires that the network media be held, the DVA MUST hold the media stream from it to the network by use of the held media procedures defined in [RSTF].

When service logic requires that currently held network media be resumed; the DVA MUST retrieve the media stream from it to the network by use of the media retrieval procedures defined in [RSTF].

### 6.2.4 Announcements

Users are alerted to some network conditions or states by announcements. Announcements could be informational tones or messages played to end users in response to events that occur in the network. An announcement could be stored or generated locally by the DVA to the end user, or could be received by the DVA as an audio from a network-based media server.

The DVA MUST follow [RSTF] in creating announcements. The DVA MUST send the created announcement to the PP as in-band media.

### 6.2.5 In-Service and Out-of-Service States

[DECT-HDV] specifies the procedure to proactively advertise to the DECT handsets that network connectivity has been lost. This procedure allows a message to be displayed on the handsets to inform the user that the network is down before the user goes off-hook.

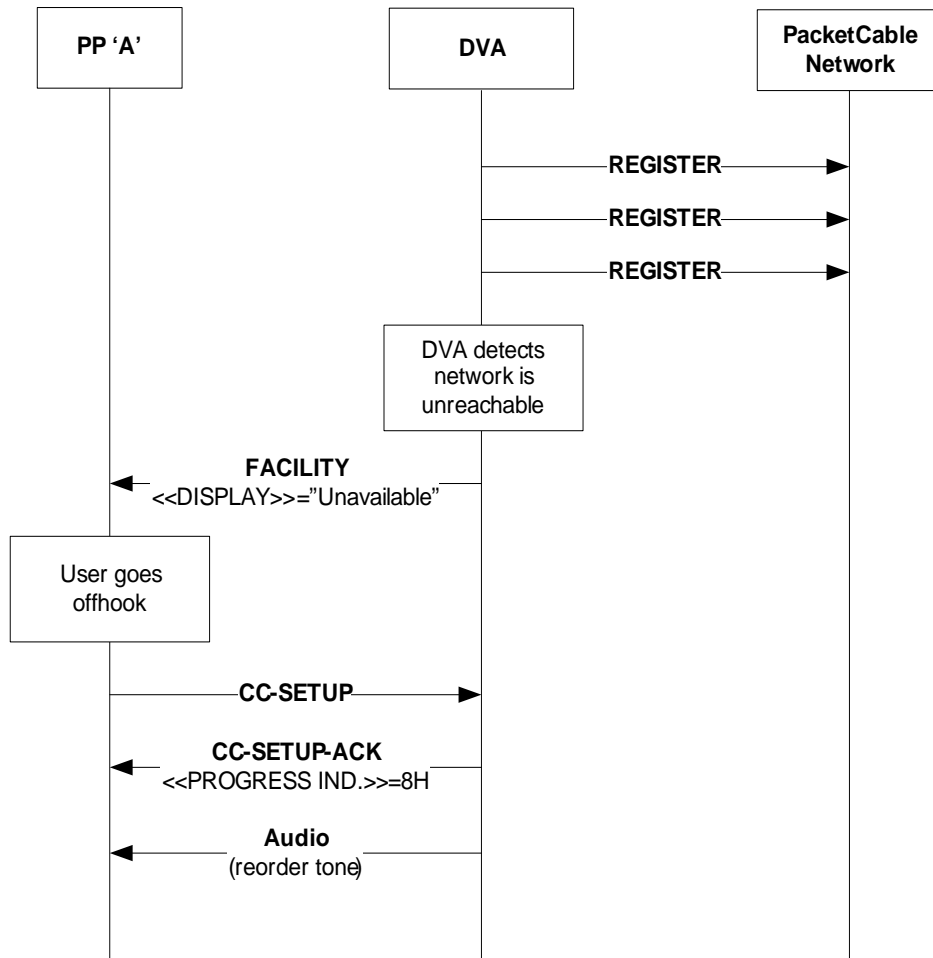
As specified in [RSTF], the DVA is required to inhibit dial tone presentation to the user going off-hook when the STUN server can not be reached but network connectivity is present. The DVA is required to present a reorder tone to the user going off-hook when network connectivity has been lost (the last REGISTER transaction has failed).

If the DVA receives a CC-SETUP message from the PP to initiate an external call after the most recent registration attempt failed, the DVA MUST complete the CC-SETUP transaction normally and present a reorder tone to the PP.

If the DVA receives a CC-SETUP message from the PP to initiate an external call after the most recent registration attempt succeeded but the most recent STUN transaction failed, the DVA MUST complete the CC-SETUP transaction normally and refrain from presenting dial tone to the user.

In both the cases, the DVA MUST inform the PP user about this condition by following the procedures in [DECT-HDV] for Service Status.

An example call flow for this scenario is shown in Figure 8.



**Figure 8 - Loss of Network Connectivity produces reorder tone**

### 6.2.6 Early Media

Early media is the media exchanged before the call is fully established. The media stream may be one-way or two-way.

The usual example of one-way media is the ring back when the DECT PP A calls network hosted UE B and A hears ring back tone from B via early media even before the network side SIP dialog is confirmed (which happens when B answers the call). An example of two-way media is PIN entry: the DECT PP A calls network hosted UE B that has Call Blocking services, and A is connected to a media server for collection of a PIN. The media stream is bi-directional; one stream to the media server for PIN digit reception, the other from the media server to request the entry of the PIN. In both examples the call is not considered established as the SIP dialog initiating transaction is not complete.

As specified in [DECT-HDV], the DVA is required to open the U-plane as part of the CC-SETUP transaction. As the U-Plane is connected immediately and prior to reception of any SIP response by the DVA, the application of audible ring back and any network-provided early media is entirely handled by the DVA with no additional signaling required on the DECT interface.

The DVA performs the following actions upon reception of SIP responses during call setup:

- Upon reception of an 18x response containing SDP, the DVA MUST through-connect received RTP packets that conform to the media description provided by the SDP.
- Upon reception of an 18x response containing no SDP, the DVA MUST provide audible ring back tone.

Figure 9 shows a representative call flow for the application of early media originating from the PacketCable network. It illustrates the procedural requirements detailed for the DVA.

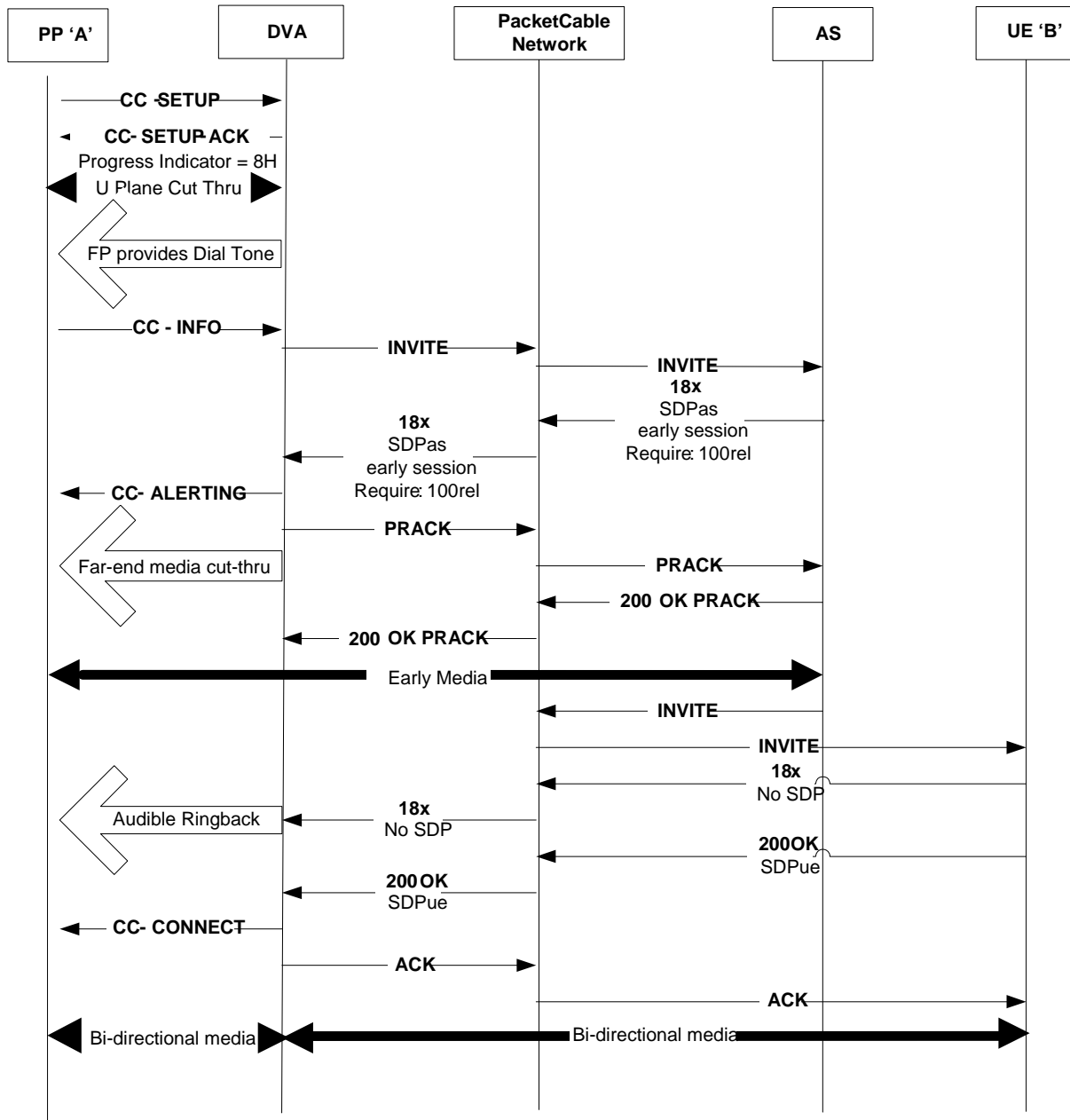


Figure 9 - Early Media

### 6.2.7 No Answer Timeout

No Answer Timeout prevents an incoming call from causing the application of power-ringing indefinitely. Specifically, No Answer Timeout causes an inbound call to be rejected once the call has been offered and unanswered for a time exceeding the No Answer timer duration.

The behavior of this feature is triggered by timer expiration on the DVA. Upon expiration of the timer, the DVA terminates the inbound call leg by following the requirements in [RSTF]. Furthermore, upon expiration of the timer, the DVA MUST also terminate the call over the DECT interface following procedures in [EN 300 444] and [DECT-HDV].

## 6.3 Residential SIP Telephony Features

This section specifies DECT to SIP mapping requirements for the telephony features specified in [RSTF].

### 6.3.1 Features with no additional DECT-SIP mapping requirement

The following [RSTF] features either do not involve the mapping between SIP messages and DECT messages, or the involved mapping is already covered by other sections of this specification, such as Section 6.1 and Section 6.2.

- Caller ID Per-Line Blocking
- Caller ID Per-Call Blocking (CIDS-Suppression)
- Caller ID Per-Call Delivery (CIDS-Delivery)
- Caller ID with Call Waiting (CIDCW)
- Anonymous Call Rejection
- Call Forwarding Don't Answer
- Call Forwarding on Busy Line
- Remote Activation of Call Forwarding
- Call Forwarding to Voice Mail
- Outbound Call Blocking
- Solicitor Blocking
- Collect Call blocking
- Operator Services
- Subscriber Programmable PIN
- Speed Dialing
- Customer-Originated Call Trace
- Screen List Editing

### 6.3.2 Caller ID

Caller ID Display is a feature that allows a subscriber to see who is calling before answering the call. Caller ID Display consists of two sub-features: Calling Number Display (CND) and Calling Name Display (CNAMD).

For an incoming call, the Caller ID information, if available, is carried in the P-Asserted-ID header of the INVITE request. The P-Asserted-ID may not be available if the caller has requested the call to be anonymous. The DVA can convey the Caller ID information as <<CALLING PARTY NUMBER>> and <<CALLING PARTY NAME>> in the CC-SETUP or IWU-INFO message sent to the PP as specified in [DECT-HDV]. If the call time is available, the DVA sends the call time as a <<TIME-DATE>> IE in an IWU-INFO message per [DECT-HDV].

If the CND feature is activated, the DVA MUST set <<CALLING PARTY NUMBER> by following the requirements in [RSTF] for constructing the calling party number to be displayed.

If the CNAMD feature is activated, the DVA MUST set <<CALLING PARTY NAME>> by following the requirements in [RSTF] for constructing the calling party name to be displayed.

The DVA MUST follow these Caller ID requirements in conjunction with the requirements in Section 6.3.6 for a call waiting scenario.

### 6.3.3 Call Forwarding Variable

Call Forwarding Variable (CFV) is a feature that allows the subscriber to activate forwarding of all calls that go to the subscriber's public identity to another location. The forward-to address can be provided through subscriber interaction with the CFV feature, or the network operator can configure the forward-to address for a subscriber. With CFV, the forwarding happens before the DVA alerts (rings) the called party, so the forwarded public identity has no opportunity to answer the call.

When CFV is successfully activated or deactivated, the DVA MUST notify the subscriber by generating and transmitting an in-band confirmation tone to the PP on which the feature is being activated or deactivated.

With CFV active, upon receiving a SIP NOTIFY message from the network about newly forwarded calls, the DVA MUST:

- Apply a ring splash to the PP, by following the ring generation requirements in [DECT-HDV].
- Send a FACILITY message to the PP with a <<DISPLAY>> IE that conveys the provisioned display string for newly forwarded calls defined in Table 1, if it is not a null string.

If the special conditions dial-tone is enabled for the subscriber with the CFV feature activated, then each time a subscriber with CFV activated goes off-hook, the DVA MUST play a locally-generated special conditions dial-tone as a reminder that CFV is active, by following the tone generation requirements in [DECT-HDV]. In this case, the DVA MUST also send a FACILITY message to the off-hook PP with a <<DISPLAY>> IE that conveys the provisioned display string for activation status defined in Table 1, if it is not a null string. If there are other active features that require their statuses to be conveyed to the PP as well, the DVA MUST concatenate all status-display strings (with appropriate spaces) and send the resulted string with a <<DISPLAY>> IE in a single FACILITY message.

**Table 1 - CFV Display Strings**

Data	Type	Persistence	Default Value	Range	Scope	Provisioning Requirement
CFV display string for newly forwarded calls	String	Volatile	Null string	N/A	Per FP	Mandatory Read-Write
CFV display string for activation status	String	Volatile	Null string	N/A	Per FP	Mandatory Read-Write

An example call flow for feature activation is shown in Figure 10 .

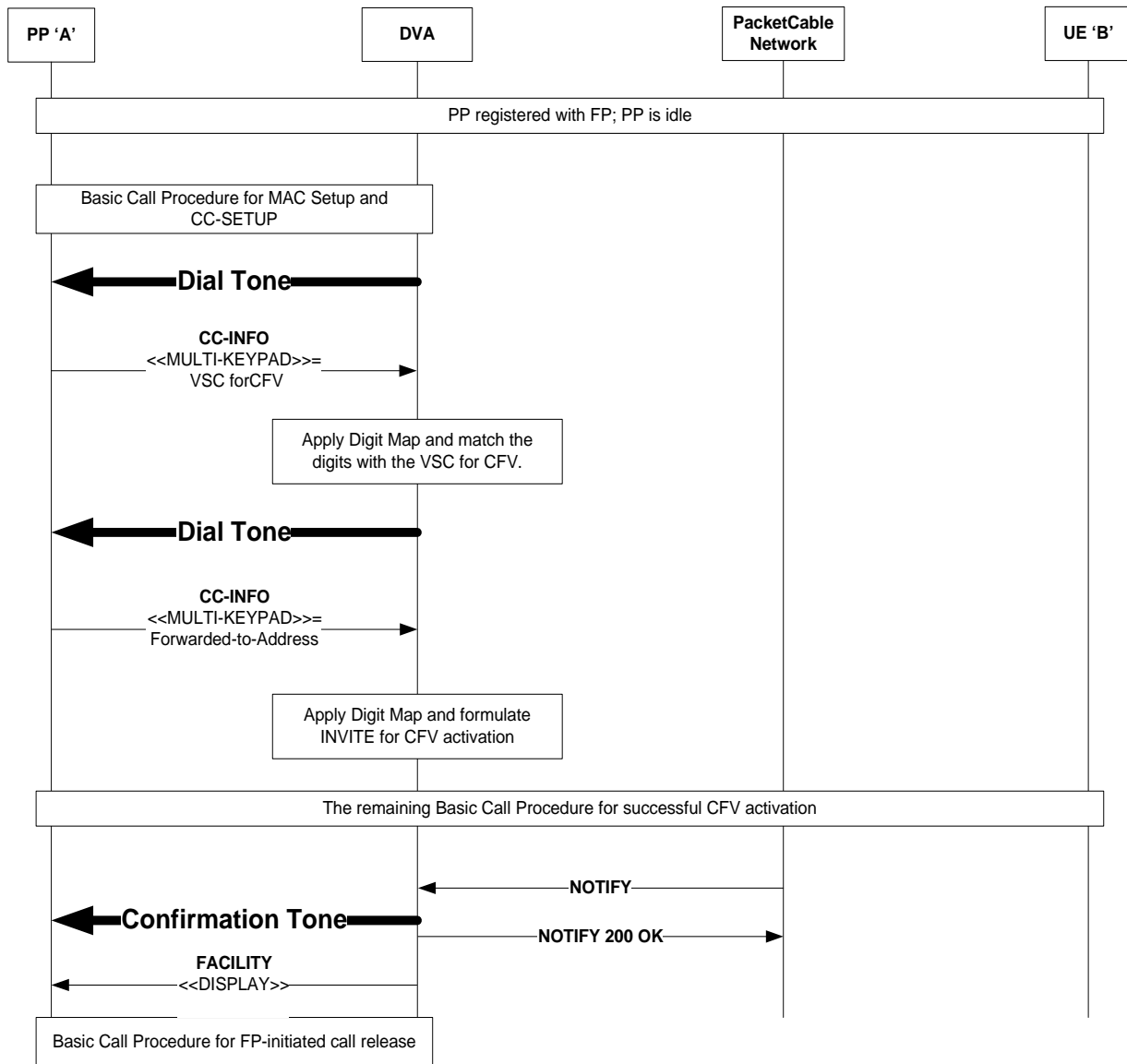
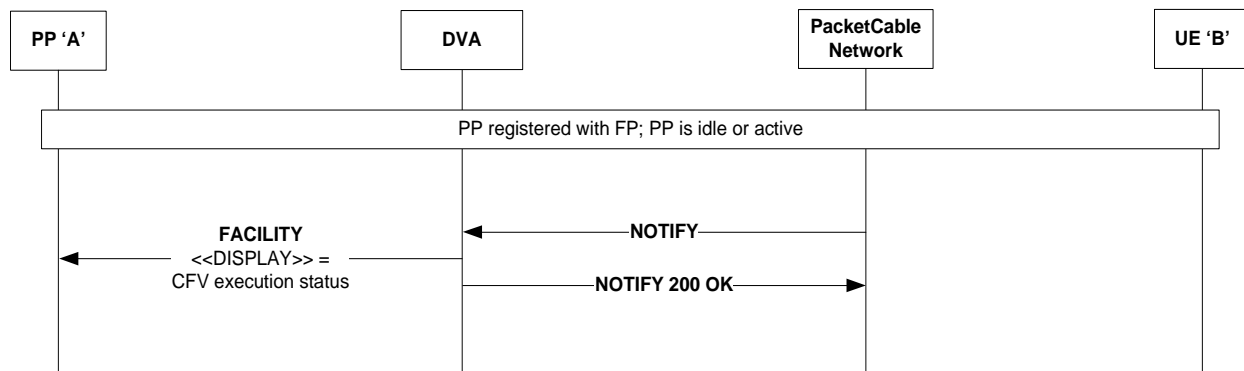


Figure 10 - Successful CFV Activation

An example call flow for status change notification is shown in Figure 11.



**Figure 11 - CFV Status Change**

### 6.3.4 Selective Call Forwarding

Selective Call Forwarding (SCF) is an incoming call management feature that allows the subscriber to define a special list of addresses and remote addresses. Incoming calls from public identities that are on the list are forwarded to the remote address. With SCF, the forwarding happens immediately and the original target public identity has no opportunity to answer the call.

With SCF active, upon receipt of a SIP NOTIFY message from the network about newly forwarded calls, the DVA MUST:

- Apply a locally-generated ring splash to the PP, by following the ring generation requirements in [DECT-HDV].
- Send a FACILITY message to the PP with a <<DISPLAY>> IE that conveys the provisioned SCF display string for newly forwarded calls defined in Table 2, if it is not a null string.

**Table 2 - SCF Display String**

Data	Type	Persistence	Default Value	Range	Scope	Provisioning Requirement
SCF display string for newly forwarded calls	String	Volatile	Null String	N/A	Per FP	Mandatory Read-Write

### 6.3.5 Hook-Flash Processing

Several telephony features allow the subscriber to use hook-flash to trigger feature logic. The hook-flash in turn prompts the PP to send to the DVA a CC-INFO message with a <<KEYPAD>> or <<MULTI-KEYPAD>> IE that has a value for Register Recall, as described in [DECT-HDV].

The DVA interprets the Register Recall indication as a hook-flash as described in [DECT-HDV], and MUST follow [RSTF] for the requirements related to hook-flash processing.

### 6.3.6 Call Waiting

The Call Waiting (CW) feature provides the subscriber with the ability to detect an incoming call while in an existing call, and to answer that call by hook-flashing, putting the existing call on hold.

The CW logic is provided by the DVA, by following the requirements in [RSTF]. The subscriber uses hook-flash to switch between the active and the waiting calls. The DVA follows requirements in [DECT-HDV] for hook-flash processing. The DVA follows [RSTF] for generating call waiting signal. If the call waiting signal is a call waiting

tone, the DVA MUST apply the tone to the PP, by following the tone generation procedure in [DECT-HDV]. The DVA follows Section 6.3.2 for the mapping requirements for Caller ID with Call Waiting.

An example call flow for Call Waiting is shown in Figure 12.

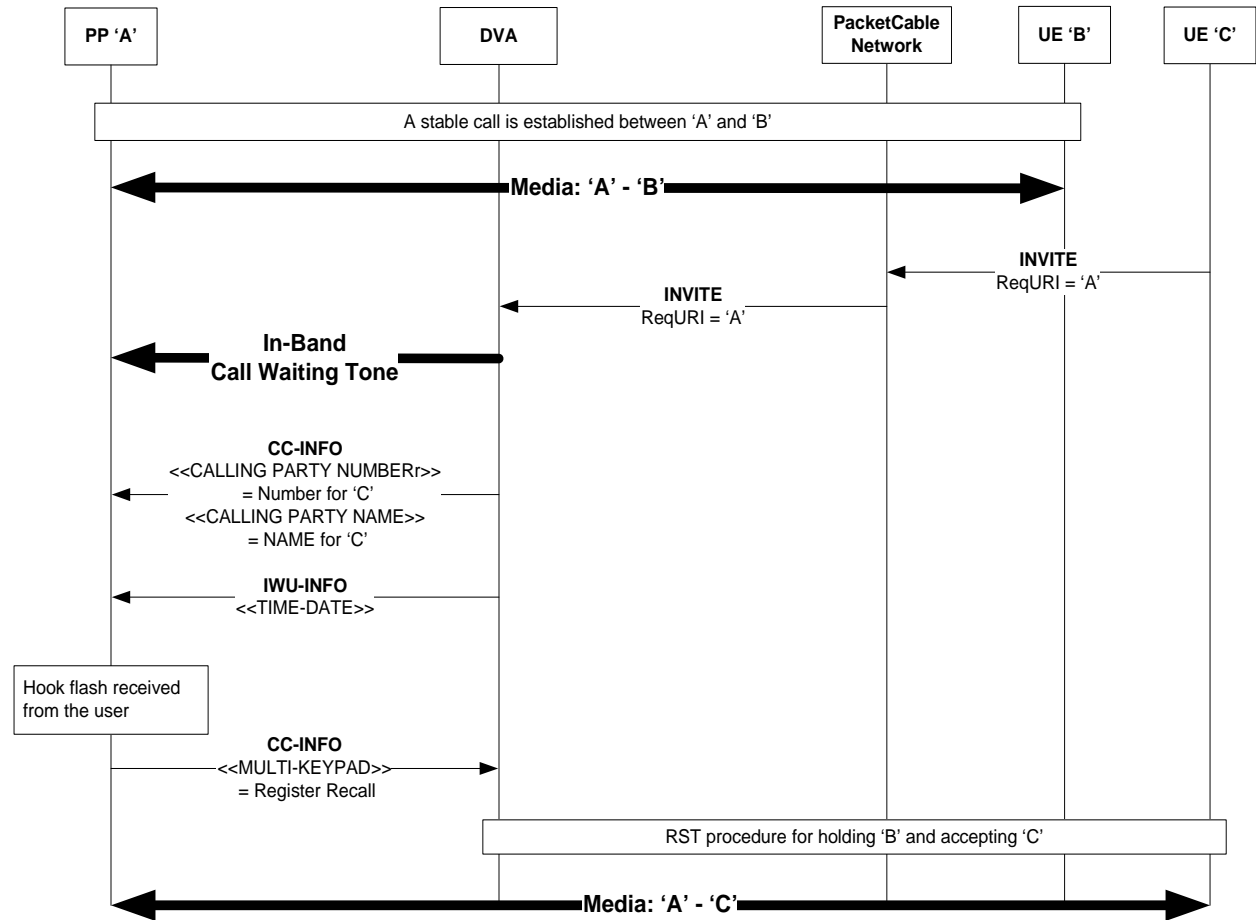


Figure 12 - Call Waiting

### 6.3.7 Call Hold

During an active call, the Call Hold feature allows the user to place the call on hold, without ending the call. The call is kept on hold, allowing the user to initiate other calls without impacting the call state of the held call.

Activation of Call Hold is achieved by the user entering a hook-flash, receiving recall dial tone and dialing the defined VSC for Call Hold. Similarly de-activation (retrieval) of a call previously held by this mechanism is achieved by the user entering a hook-flash followed by the same VSC (i.e., the VSC acts as a toggle). The requirements for handling a hook-flash are defined in Section 6.3.5. The requirements for handling the entry of the VSC are defined in Section 6.2.2.

On reception of the VSC for Call Hold for a currently active call, the DVA UE MUST hold the network media using the mechanisms defined in Section 6.2.3. On reception of the VSC for Call Hold for a currently held call, the DVA MUST retrieve the network media using the mechanisms defined in Section 6.2.3.

All other procedural requirements pertaining to a call that is held by this mechanism are defined in [RSTF].

### 6.3.8 Call Transfer

Call Transfer is the transfer of one caller (the transferee) to another party (the transfer-to party), in which the transfer is initiated by the originator of the second call (the transferor). There are two common varieties of Call Transfer - consultative and blind. These are defined in [RSTF] in terms of the state of the second call: if the call is unanswered before completion of the transfer sequence, the transfer is considered blind; otherwise, it is consultative.

The DVA follows the procedures in [RSTF] for handling Call Transfer. User interaction with the service logic is via the use of hook-flash to both enable the initiation of a second call leg when on an active call, and subsequently to transfer the two remote parties. The requirements for handling a hook-flash are defined in Section 6.3.5.

#### 6.3.8.1 Scenario 1: DECT PP Initiated Transfer

A DECT PP that is already part of an existing two-party call upon detection of hook-flash issues a CC-INFO as defined in Section 6.3.5. The DVA SHOULD hold the currently active call using the procedures defined in Section 6.2.3. The DVA then provides dial-tone to the PP and MUST follow basic call procedures (Section 6.1) to initiate a second call.

Completion of the transfer occurs when the PP goes on-hook and issues a CC-RELEASE to the FP. If the DECT PP goes on-hook while the second call attempt is still alerting, then this is termed a blind transfer. If the DECT PP goes on-hook after the second call attempt has been established, then this is termed a consultative transfer. On detection of the CC-RELEASE on the second call attempt, the DVA MUST make use of the procedures defined in [RSTF] to complete the call transfer via use of the REFER method. On reception of a NOTIFY indicating that the transfer was successful, the DVA MUST release all call legs on the PacketCable network.

Figure 13 shows a representative flow for the case where the transferor is a DECT PP:

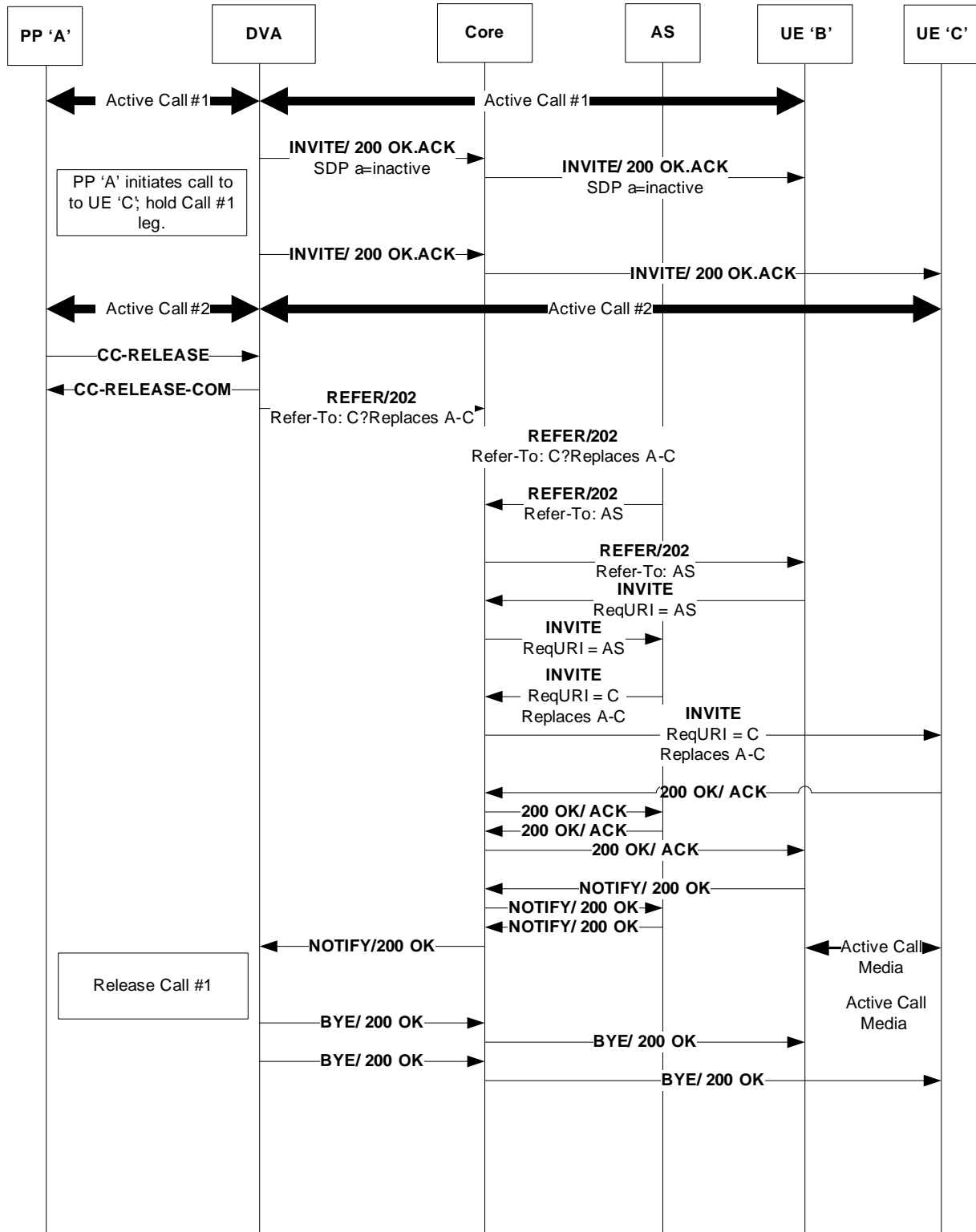


Figure 13 - PP initiated Call Transfer

**6.3.8.2 Scenario 2: Remote Endpoint Initiated Transfer**

The DVA that receives a REFER in accordance with [RSTF] follows the procedures defined therein for completion of Call Transfer. The DVA that receives an INVITE containing a Replaces header in accordance with [RSTF] follows the procedures defined therein for handling of the INVITE.

The DVA MAY make use of the CC-INFO message with <<CALLING PARTY NUMBER>> and <<CALLING PARTY NAME>> IEs to update the connected or calling party information presented to DECT PPs involved in the call transfer as either transferee or transfer-to. Updated calling or connected party information is received in a P-Asserted-Identity header either in an INVITE (Replaces) received by the DVA or in the 200 OK to an INVITE set by the DVA as a result of the reception of a REFER.

Figure 14 shows a representative flow for the case where the transferor is in the PacketCable network.

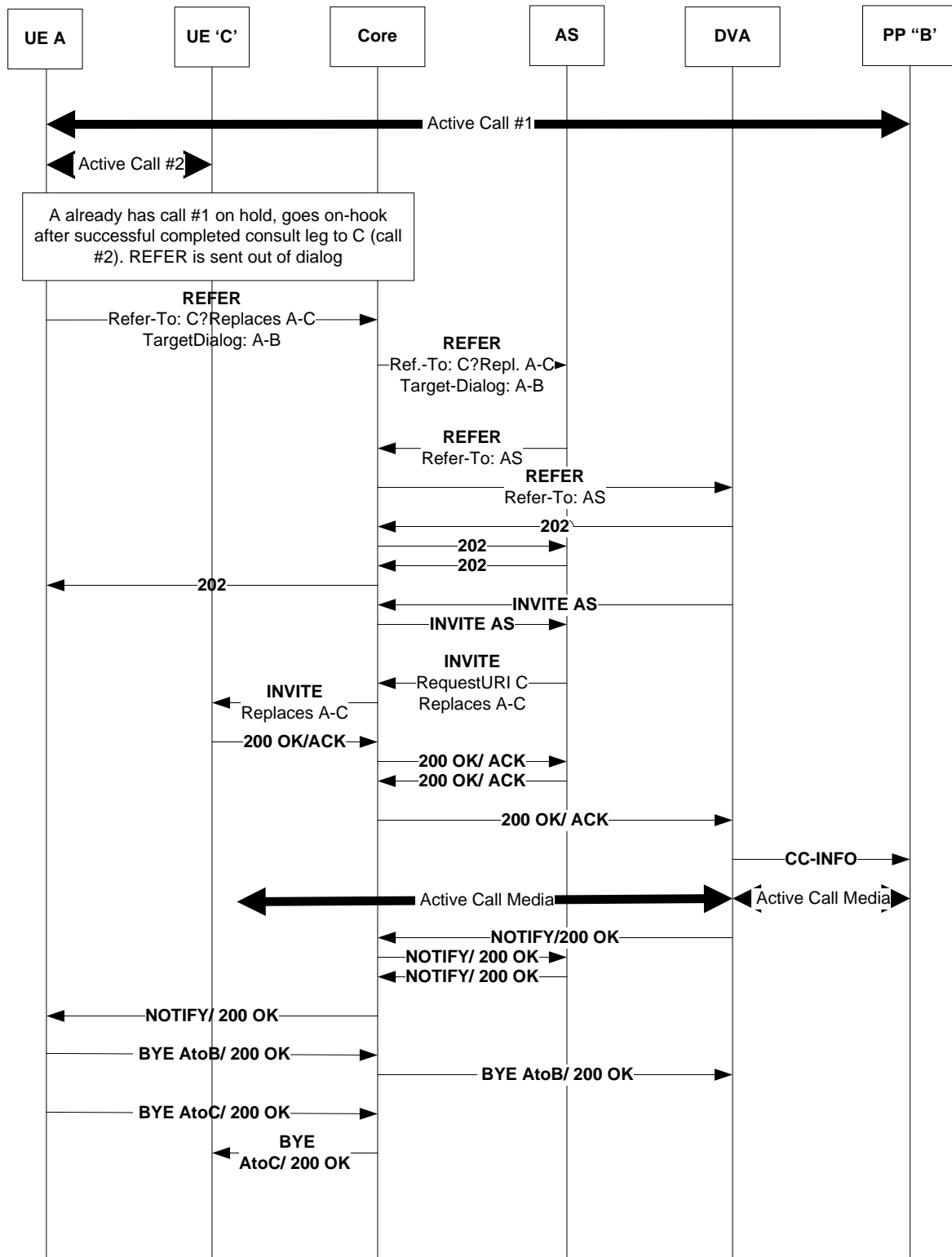


Figure 14 - Remote endpoint initiated Call Transfer

### **6.3.8.3 Message Mapping**

This mapping applies to the reception at the DVA of an INVITE containing a Replaces header identifying an existing SIP context on the DVA. Media bridging will occur at the device. However, the DVA MAY map the P-Asserted-Identity in the INVITE to a <<CALLING PARTY NUMBER>> and <<CALLING PARTY NAME>> IEs in the CC-INFO message to the DECT PP that is part of the existing call identified by the Replaces header. Similarly, on reception of a 200 OK for an INVITE sent by a DVA while handling the reception of a REFER, the DVA MAY map the 200 OK to a CC-INFO to the DECT PP that is part of the call dialog associated with the REFER.

### **6.3.9 Three-Way Calling**

Three-Way Calling (3WC) allows a subscriber to add a third party to an existing two-party active call without operator assistance.

The 3WC logic is provided by the DVA, by following the requirements in [RSTF]. While in a stable call, the subscriber uses hook-flash to initiate the call to the third party. The DVA follows requirements in Section 6.3.5 for hook-flash processing. The DVA MUST apply a locally-generated dial tone to the PP, by following the tone generation procedure in [DECT-HDV]. After the dial tone is applied, the DVA follows [RSTF] for processing the third-party address entered by the subscriber and for initiating the call to the third party. Upon connecting to the third party, the subscriber performs hook-flash again to enter the 3-way call.

Figure 15 shows a representative flow for 3WC.

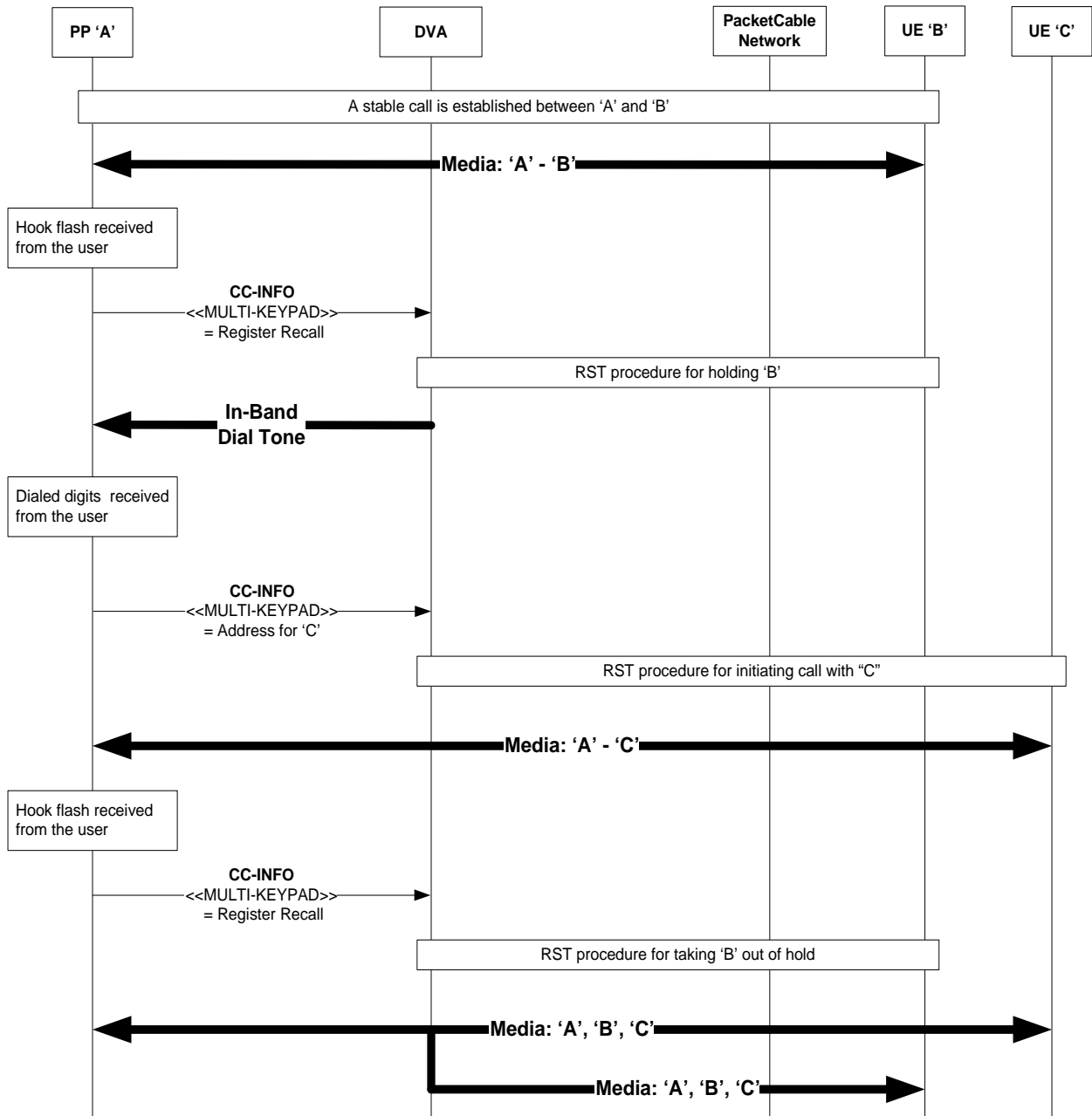


Figure 15 - Three Way Call

### 6.3.10 Call History

Call History consists of two discrete services, Automatic Recall and Automatic Callback.

The Automatic Recall (AR) service allows a user to automatically return a call to the last calling address, i.e., the entity that terminated a call to this UE, whether the call was answered or not. The AR service works even when the last call received by the user was anonymous.

The Automatic Callback (AC) service allows a user to automatically call back the last called address, whether the call was answered by the called party or not. The AC service works even when the last call made by the user was to an address not known by the subscriber.

Activation of either AC or AR is achieved by the user dialing the defined VSC for the requested service; the requirements for handling the entry of the VSC are defined in Section 6.2.2.

On reception of the VSC for AC or AR, the DVA follows the procedures defined in [RSTF] for initiating an AC or AR call attempt and the subsequent handling of SIP responses from the network.

If the DVA receives a 486 User Busy response to the AC or AR call attempt, then the DVA MUST NOT render any call failure tone or announcement to the PP; instead, the DVA follows the procedures defined in [RSTF] for subscription to the dialog state of the target of the call attempt.

The DVA makes use of Third Party Call Control (3PCC) techniques to connect the PP and the AC or AR target once the AC or AR target becomes idle as defined below:

- If the DVA receives a NOTIFY message from the network on the dialog created by the subscription to the dialog event package indicating that the AC or AR target is now idle, and the PP that initiated the AC or AR attempt is also idle, then the DVA MUST send a CC-SETUP to the PP including a <<SIGNAL>> IE populated as defined in [DECT-HDV] to provide special alerting at the PP.
- The DVA MUST start a timer for the special alerting duration and handle timer expiry as defined in [RSTF].
- Upon reception at the DVA of a CC-CONNECT indicating that the PP has answered the call attempt to it, the DVA MUST send an INVITE to the target of the AC or AR attempt as defined in [RSTF].
- The DVA, on reception of an 18x progress message from the network with no SDP, MUST provide local ring back tone to the PP.
- The DVA, on reception of an 18x progress message from the network with SDP, MUST cut-through the far end media to the PP.
- Once the AC or AR target answers the call attempt, the DVA MUST NOT map the SIP 200 OK to a CC-CONNECT as per basic call procedures (Section 6.1); instead, the DVA MUST cut-through far end media if it has not done so already.

Figure 16 shows a representative call flow for the invocation of Automatic Recall. Note that Automatic Callback exhibits the same flow, except that the Request URI in the initial INVITE will be that of the last party called.

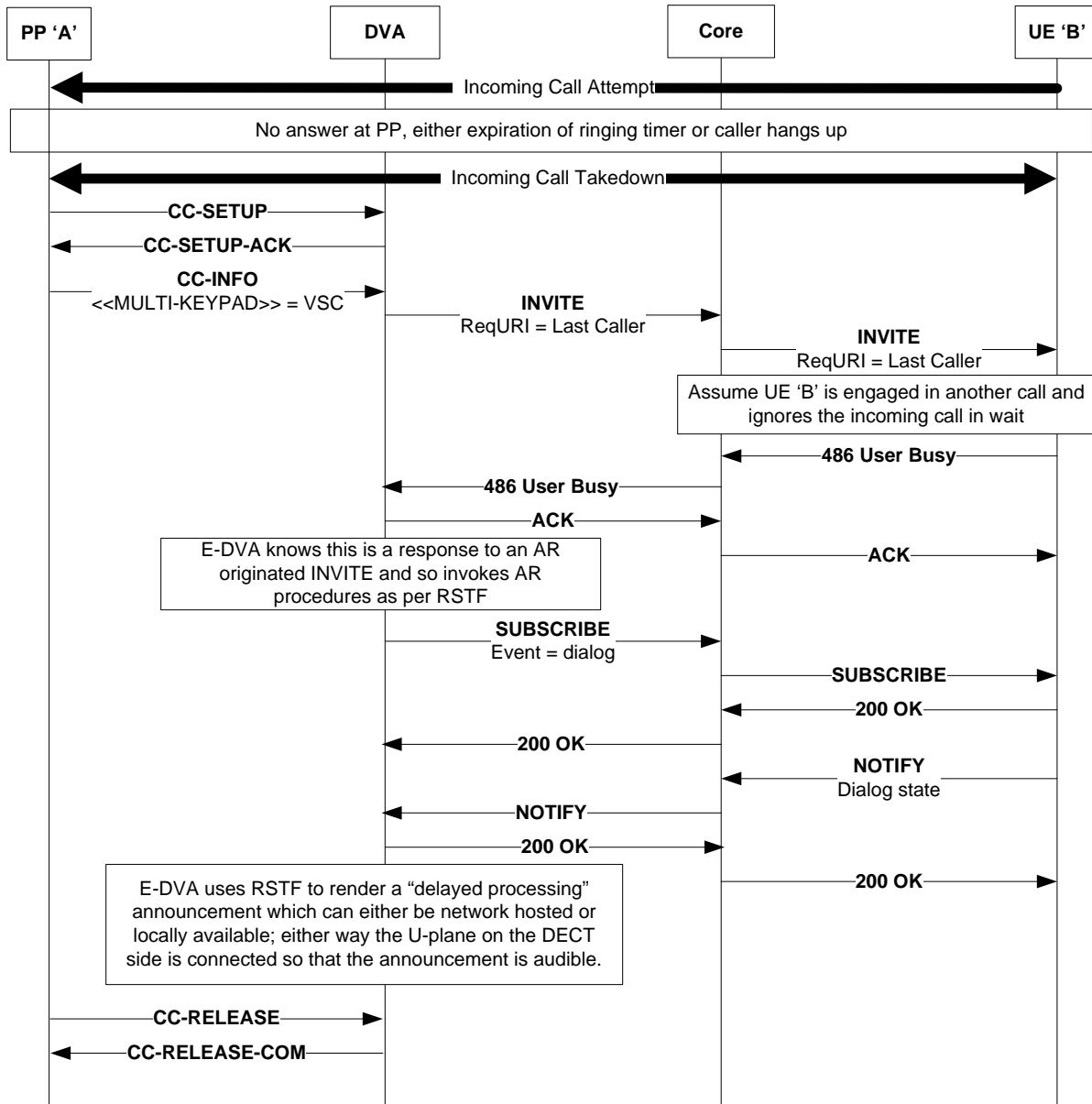


Figure 16 - Auto Recall - Invocation

Figure 17 shows a representative call flow for the continuation of an Automatic Recall attempt once the target becomes idle. Note again that Automatic Callback exhibits the same flow except that the Request URI in the INVITE will be that of the AC target.

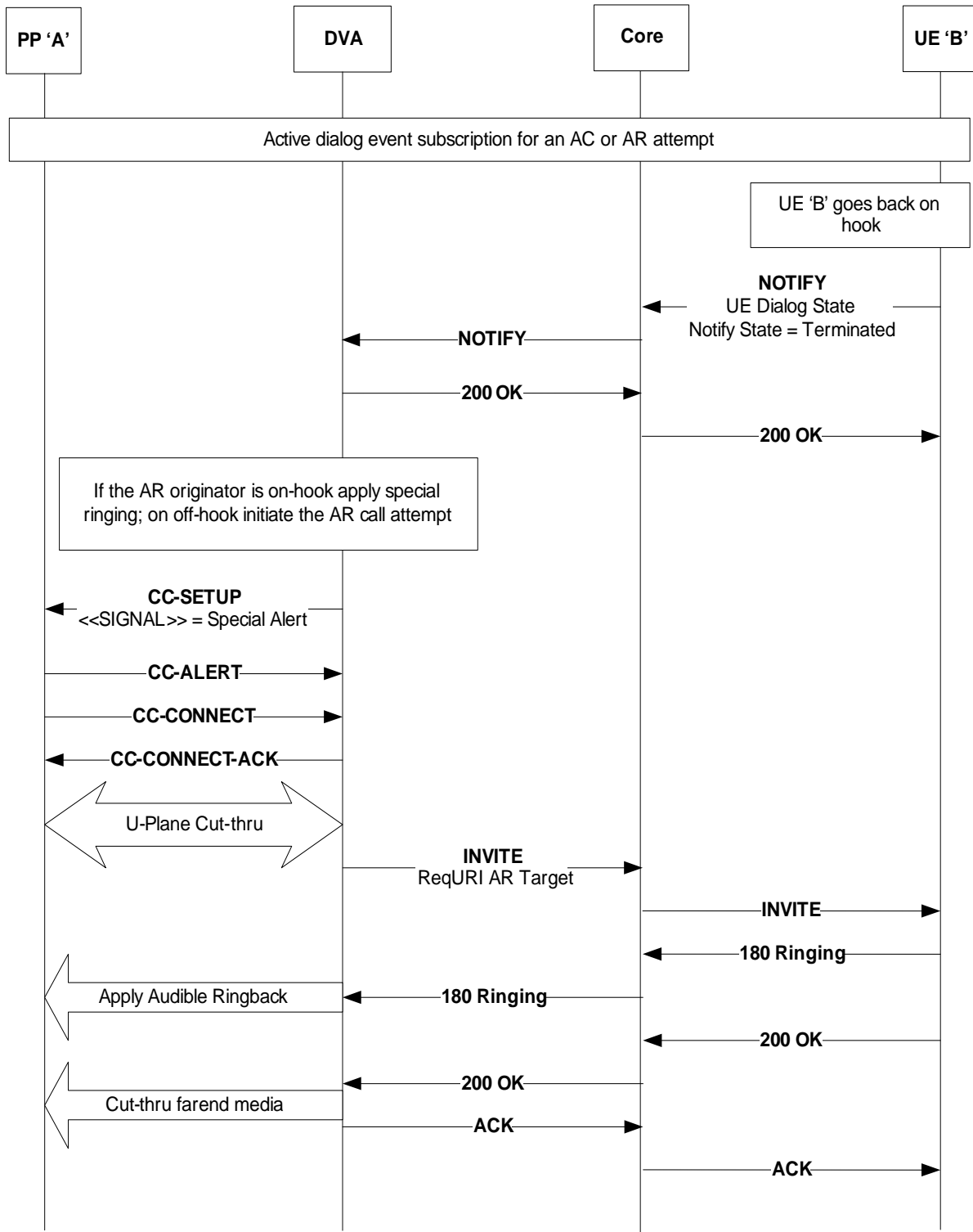


Figure 17 - Auto Recall - Target becomes idle

### 6.3.11 Emergency Services

The Emergency Services allow the subscriber to initiate emergency calls from a PP. The detailed description and general assumptions for the emergency-call support on a DVA are provided in [RSTF].

With the PP registered with the DVA, the subscriber makes an emergency call by entering the digit string designated for the emergency services. The DVA applies the digit map to process and recognize the designated digit string, and subsequently to carry out the emergency call procedure as specified in [RSTF]. The DECT-SIP mapping requirements for digit map processing are provided in Section 6.2.1.

The DECT setup procedure for an emergency call is the same as that for a basic call, which is described in Section 6.1. As specified in [RSTF], there are several special requirements related to the maintenance of an emergency call. In particular, when the subscriber goes on-hook during an emergency call, the DVA performs Network Hold on the call. The DECT-SIP mapping requirements for Network Hold are provided in Section 6.3.7.

### 6.3.12 Do Not Disturb

The Do-Not-Disturb (DND) feature allows the subscriber to put a DVA into a mode in which the DVA does not send an alert when a call is presented.

The subscriber can activate or deactivate the DND feature by entering a Vertical Service Code (VSC). In this case, the DVA follows Section 6.2.2 for the DECT-SIP mapping requirements associated with VSC processing.

Upon completion of activation or deactivation operation, the DVA is notified by the network about the status change of the feature. When the NOTIFY message is received, the DVA MUST apply the confirmation indicator (a tone or an announcement) to the PP as in-band media.

With the DND feature active and upon detection of off-hook condition, the DVA MUST follow the [DECT-HDV] tone generation procedures to apply a recall dial tone to alert the subscriber that the DND is still active, as described in [RSTF]. In this case, the DVA MUST send a FACILITY message to the PP with a <<DISPLAY>> IE that conveys the provisioned display string defined in Table 3, if it is not a null string. If there are other active features that require their statuses to be conveyed to the PP as well, the DVA MUST concatenate all status-display strings (with appropriate spaces) and send the resulting string with a <<DISPLAY>> IE in a single FACILITY message.

**Table 3 - DND Display String**

Data	Type	Persistence	Default Value	Range	Scope	Provisioning Requirement
DND display string for activation status	String	Volatile	Null string	N/A	Per FP	Mandatory Read-Write

### 6.3.13 Distinctive Alerting

The Distinctive Alerting (DA) feature allows different alerting patterns to be rendered to the subscriber on a per-call basis. The selection of a particular alerting pattern for a given incoming call is determined by the DA application server in the terminating network.

The subscriber can program the DA feature by entering a Vertical Service Code (VSC). In this case, the DVA follows Section 6.2.2 for the DECT-SIP mapping requirements associated with VSC processing.

Upon receiving an incoming INVITE from the network, the DVA determines the alert signal to be applied to the subscriber by mapping the Alert-Info header in the INVITE message, as described in [RSTF]. If the resulting alert signal is a ringing signal, the DVA MUST follow [DECT-HDV] in applying the ringing signal to the PP.

### 6.3.14 Message Waiting Indicators

Message Waiting Indicators (MWI) are used by the DVA to alert the PP that the status of a message account associated with the subscriber is changed. Once received, the MWI is presented by the PP to the subscriber as a visual signal and/or an audible signal.

The MWI Application Server detects the status change of the message account, and informs the DVA about the change via a SIP NOTIFY message that contains the information about the MWI event [RSTF].

Upon receiving a NOTIFY message for the MWI event, the DVA MUST send a FACILITY message to the PP with an <<EVENTS-NOTIFICATION>> IE that indicates voice message waiting, as specified in [DECT-HDV]. The DVA MUST set the Event Multiplicity value in the <<EVENTS-NOTIFICATION>> IE to be the sum of the new message count and the old message count indicated in the SIP NOTIFY message.

Figure 18 shows a representative call flow for MWI.

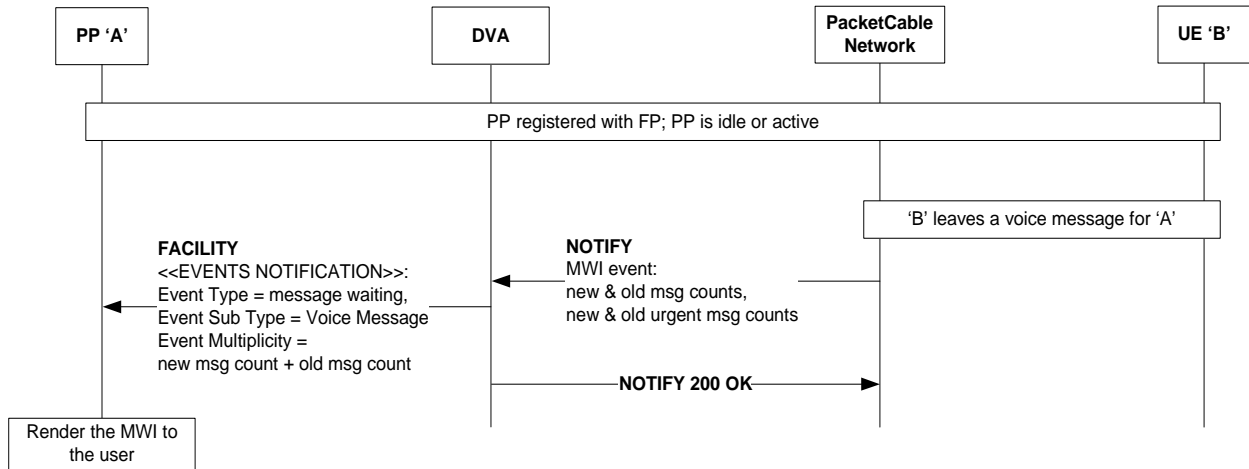


Figure 18 - Message Waiting Indicator

## **Annex A Device Requirements (Normative)**

### **A.1 Embedded Device (E-DVA) Requirements**

The E-DVA MUST conform to the DVA requirements in this specification.

The E-DVA MUST conform to [DECT-HDV] for requirements pertaining to the DECT interface and its interactions with the analog phone interface.

The E-DVA MUST conform to the requirements in [E-DVA] with the following additions:

- For codecs over the PacketCable network, the E-DVA MUST support G.722 in addition to G.711.
- The E-DVA MUST support the additional provisioning requirements in [DECT-PROV].

For loopback testing, the loopback points within the E-DVA are the same for both the DECT interface and the analog interface. Therefore, these loopback tests do not reach over to the PP.

### **A.2 Standalone Device (S-DVA) Requirements**

The S-DVA MUST conform to the E-DVA requirements in Section A.1, with the following exception - for NAT/Firewall traversal, the S-DVA MUST support the SIP Outbound and the ICE methodology described in [PKT 24.229].

## Appendix I      Acknowledgements

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