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1. Introduction

The *DM3 IPLink User’s Guide* provides the information necessary to build an Internet Telephony application using Dialogic Corporation’s DM3 IPLink platform.

Topics discussed in this introduction include:

- About This Guide
- IP Telephony Overview
- What You Need to Know To Create An Application
- What Resources are Available?

The *DM3 IPLink User’s Guide* assumes that the hardware is installed and the firmware has been downloaded, and therefore will not describe hardware installation or firmware download. Refer to the *DM3 IPLink Series Quick Install Card* included with each board for a description of hardware installation and Installing and Configuring The DM3 IPLink SDK for Windows NT for firmware download.

1.1. About This Guide

This document is divided into two sections:

- **Section 1** provides an introduction to the DM3 IPLink platform. It includes:
  - Chapter 2. The DM3 IPLink Platform
  - Chapter 3. Host Application

- **Section 2** provides programmer resource material. It includes:
  - Chapter 4. Programming Model
  - Chapter 5. Using the NetTSP Resource
  - Chapter 6. Setting the DM3 IPLink Parameters
  - Chapter 7. Debugging
Background information about Internet Telephony can be found in the Appendix.

**NOTE:** Dialogic Corporation recommends that first-time users read through this entire User’s Guide before beginning to write a host application. Experienced users can refer directly to the chapters in Section 2 when writing an application.

### 1.2. IP Telephony Overview

Transporting voice over the Internet Protocol (IP) requires computationally intensive signal processing functions, including low bandwidth coders, advanced echo cancellation, multimedia call control and high performance protocol stacks.

The DM3 IPLink platform allows a “voice over IP” call to be connected via the DM3 platform to the SCbus. By adding additional SCbus boards, one can build a variety of applications, such as an IP—PSTN gateway.

The DM3 IPLink board and firmware, together with standard Dialogic Corporation SCbus PSTN cards, provide all the call control functions necessary for IP telephony:

- Answer incoming calls from either the IP network or the PSTN network
- Collect relevant information from the incoming call
- Make a corresponding call on the complementary network
- Bridge the two calls

This capability can be used to build a gateway, bridging the traditional circuit-switched telephony world with the Internet.

The IP Telephony configuration enables current Dialogic Corporation telephony resources to be mixed and matched with an IP connection and Dialogic’s DM3 IPLink platform to build a wide variety of solutions.
1. Introduction

1.3. What You Need to Know To Create An Application

This section describes the knowledge you will need to write an Internet telephony application. Before beginning to write an application, you should already be familiar with:

- C or C++ programming language
- Windows NT Programming

In order to write an Internet telephony application, you should be familiar with:

- DM3 messages, clustering, timeslot assignment
  - See DM3 Architecture Overview for a description of clustering concepts and timeslot assignment
  - See IPLink Reference Guide for a complete description of TSC messages
    - For a list of TSC messages - see Section 5.1.2. TSC Messages Used by the NetTSC Component
  - See DM3 Standard Component Interface Messages for a complete description of DM3 standard messages
    - For a list of standard messages - see Section 5.1.1. Standard Messages Used by the NetTSC Component
- Dialogic System Software for Windows NT (when using Dialogic Corporation SCbus PSTN cards)
  - The Dialogic System Software on-line bookshelf is included with this IPLink SDK.
- DM3 Direct Interface
  - See DM3 Direct Interface Function Reference and Using the DM3 Direct Interface for Windows NT
- Win32 Synchronization Methods
  - See Using NSM for Event Notification with DM3, R4, and Win32 Devices
2. The DM3 IPLink Platform

This chapter presents a high-level description of the DM3 IPLink platform, describing both the hardware platform and the software.

Topics discussed include:

- The DM3 IPLink Platform
- DM3 IPLink Capabilities
- Quality of Service
- Hardware Platform
- The NetTSP Resource
- Timeslot Management
- Line Administration Component
- Internal Components

2.1. What is the DM3 IPLink Platform?

The DM3 IPLink platform includes both hardware and software. Figure 1 displays the various parts of the IPLink platform. The grayed areas are internal components not accessible to the application writer and are shown in order to present a complete picture of the IPLink platform. The components shown in circles are not supplied by Dialogic Corporation.
2.2. The DM3 Platform

This section introduces basic concepts and terminology relating to the DM3 Mediastream Architecture. DM3 is an architecture on which a whole set of Dialogic products are built. The DM3 architecture is open, layered, and flexible, encompassing hardware as well as software components.

2.2.1. The DM3 Platform

The software on the DM3 platform consists of the following types of software:

- **Core platform software**, which includes the platform-specific DM3 kernel and device driver. All base platforms of the same type will have the same core platform software.

- **DM3 resource software**, which provides the features of a resulting product.
2.2.2. The DM3 Technology Resources

A DM3 technology resource is a set of embedded resource software that provides a specific set of features, such as voice over Internet Protocol, for a DM3 product. A resource represents a set of features to be performed and the functions of one resource are logically independent of any other resource. A DM3 resource closely matches the SCSA concept of a resource (such as a player, recorder, fax transmitter, etc.).

A DM3 signal-computing resource is made up of a host software component that resides on the host platform containing the DM3 board(s), and one or more DM3 firmware components that reside on the various processors on the boards themselves.

2.2.3. DM3 Component

DM3 components are logical entities that implement specific features within a resource. A resource may be separated into components to distribute functions among processors (for example, control functions versus signal processing algorithms), or to provide alternative or optional services within a resource (such as multiple coders in a player resource).

A component instance is an addressable unit within the DM3 software architecture; it represents a single thread of control. The DM3 system resource management and messaging services operate at the instance level. A set of component instances communicate with one another using the DM3 system messaging services. A set of component instances is usually associated with a channel of call processing.

2.2.4. Clusters

A set of component instances that share SCbus timeslots and collectively deliver a set of functions for the host controller are grouped into a cluster. Clusters control how TDM data is transferred from one component to another, and how TDM data is routed to and from the SCbus. When a DM3 component instance is allocated, it is allocated into a specific cluster. Once a cluster has been created, the DM3 kernel can control the ports to allow the resources within that cluster to share ports in an intelligent manner.
While the cluster contents can be dynamically configured at run-time, default setups are usually enough for most applications and offer a higher level granularity of access to the DM3 platform.

2.2.5. DM3 Component Communication

DM3 components have two communication mechanisms:

• Sending messages
• Sending blocks of data

A message-based protocol is used for communicating between DM3 components or component instances in the system.

Bulk data in a DM3 system is manipulated using global streams to transfer the data between processors, between the host and a component or instance, and between the SCbus and a component or instance.

2.2.6. The SCbus

The SCbus allows different SCSA compatible boards to communicate with each other. The DM3 IPLink platform allows a “voice over IP” call to be connected via the DM3 platform to the SCbus.

2.3. IPLink Platform Capabilities

The DM3 IPLink platform supplies the following capabilities:

• **Call Management** is provided by the NetTSC component which, in turn, uses an H.323 stack.

• The **RTP/RTCP** media stream is a sequence of packets using UDP (User Datagram Protocol) as transport. Packetization is handled by the resources on the IPLink board.

• The following **Voice Coders** are supported by the DM3 IPLink platform (at the time of writing this manual):
  
  • G.723.1
2. The DM3 IPLink Platform

- G.711
- GSM

**NOTE:** Call your Dialogic Corporation sales engineer for the complete list of supported voice coders.

- DTMF Detection/Generation
  The IPLink platform provides both inband and out-of-band tone detection and generation.

- Packet Loss Recovery Mechanism
  The IPLink platform includes a sophisticated algorithm for lost packet recovery.

2.4. Quality of Service

Unlike the PSTN, in which quality of service is defined as a particular level of service ("toll-like"), quality of service for voice over the Internet Protocol is defined as a continuum of levels, affected by packet delay or loss, line congestion, and hardware quality such as microphone quality.

The DM3 IPLink platform is designed to operate along the entire range of quality of service, enabling the application to retrieve information necessary for correct billing.

2.5. Hardware Configuration

The IPLink platform is available in two hardware configurations:

- DM/IPLink-T1 and DM/IPLink-E1
  - Requires a host Network Interface Card (NIC) to connect to the IP network
- DM/IPLink-T1_NIC and DM/IPLink-E1_NIC
  - Network interface is included on the IPLink board
Both models of the IPLink board connect to a separate PSTN board via the SCbus.

A typical PSTN-IP Gateway configuration consists of three parts:

- Dialogic SCbus PSTN Interface Board (PSTN Connection)
- DM3 IPLink Board (IP Network Connection)
- Network Interface Board (IP Interface) or DM3 IPLink Board with NIC

Figure 2 illustrates a possible hardware configuration using an IPLink board with a separate network connector:

![Figure 2. PSTN-IP Gateway Configuration (separate Network Interface Board)](image)

Figure 3 illustrates a possible hardware configuration using an IPLink board with a network connector:
2. The DM3 IPLink Platform

Figure 3. PSTN-IP Gateway Configuration (Network Interface on IPLink)

2.6. The Net Telephony Service Provider (NetTSP) Resource

2.6.1. NetTSP Resource Overview

The NetTSP resource provides complete embedded IP call control from within the DM3 platform. The NetTSP resource allows a host application to:

- Make a call
- Answer a call
- Manage established calls
- Gather call statistics
- Connect the IP network to the SCbus

2.6.2. NetTSP Resource Features

The NetTSP resource provides all the features necessary to rapidly build an Internet phone gateway:

- High-level API
DM3 IPLink™ User’s Guide for Windows NT

- Call control capabilities (H.323)
- RTP/RTCP protocol
- Connectivity to IP Stack
- Family of voice coders
- Timeslot management

2.6.3. NetTSP Resource Elements

The NetTSP Resource contains the following component:

- **NetTSC Component**
  
The NetTSC component acts as a liaison between the host and all the sub-components in the NetTSP resource. It is the manager for all NetTSP services on the platform.

In addition, the NetTSP resource contains these additional components whose function is transparent to the host application:

- **H.323 Component**
  
  Supplies the services required for sending voice over the Internet, e.g., call management.

- **VSR (Voice Stream Resource) Component**
  
  Performs the transcoding between PCM audio to and from the PSTN and a digitally coded data stream to and from the Internet. The VSR contains the following main features:
  
  - Voice Coders
  - Tone Generator Component
  - Signal Detector Component
  - Echo Canceler Component
  - Packet Loss Recovery Module
2. The DM3 IPLink Platform

Figure 4 shows the relationship among the IP Telephony components.

Figure 4. NetTSP Architecture - DM/IPLink-T1(E1) Board
2.6.4. NetTSC Component

The NetTSC component is the primary component of the NetTSP resource. The IPLink platform has a single NetTSC component and many instances. Each instance of the NetTSC component represents a single channel of call control.

Applications using the runtime features of the NetTSP resource communicate only with NetTSC component instances.

Runtime features are features enabled after instance initialization, such as call connection, statistic gathering, etc.
Each NetTSC instance provides complete call control for inbound and/or outbound calls on its logical channel.

The NetTSC component acts as the liaison between the host and all the components of the NetTSP resource. The NetTSC component is the manager for all the TSP services on the platform.

The host application accesses the features of the NetTSP resource by communicating with an instance of the NetTSC component. Chapter 6. Setting IPLink Parameters.

2.6.5. NetTSC Component Activities

The NetTSC component performs the following activities:

- Interfaces to the NetTSP resource
- Creates clusters and allocates instances of the VSR component to each cluster (at initialization)
- Communicates with the H.323 component
- Provides information on Quality of Service (QoS) per call
- Manages Alarm Events

2.7. Line Administration Component

NOTE: The Line Administration Component is not implemented in this version of the IPLink Development Kit.

The Line Administration component collects statistics and provides information on board-level activities. The host application can communicate directly with this component to collect the required information.

2.7.1. Line Administration Component Activities

The Line Administration component performs the following activities:

- Statistic Collection (board level)
• Number of concurrent calls
• Maximum number of concurrent calls
• Average number of lost packets
• Network Alarms
• Network disconnect

2.8. Internal Components

NOTE: This section is informational only. It describes components that operate internally and are not accessible by the application.

2.8.1. Voice Stream Resource Component (VSR)

The VSR component performs the transcoding between PCM audio to and from the PSTN and a digitally coded data stream to and from the Internet. VSR activities include:
• Encoding/Decoding (Voice Coder)
• Creating RTP packets
• Tone (DTMF) detection
• Tone (DTMF) generation
• Echo cancellation
• Lost packet recovery

The operation of the VSR resource is transparent to the user, except for optionally setting several parameters. See Chapter 6. Setting IPLink Parameters for a description of parameters that may be set by the application.

Voice Coder

The voice coder implements the bi-directional translation between a compressed speech format (on the IP network) and an 8 KHz linear stream of PCM audio samples (on the SCbus).
Key features include:

- Low bandwidth
- High voice quality (3.5 MOS* rating or better)
- Low latency

The IPLink platform supports multiple coders, allowing proprietary coders as well as new standard coders to be added to the system.

**Echo Canceler**

The echo canceler removes the echo from the far signal channel.

Key features include:

- “Echo-tail” length of up to 64 ms.
- Programmable length

**DTMF Tone Detector/Generator**

The IPLink platform provides a DTMF tone detection and suppression algorithm, and supports out-of-band as well as in-band DTMF handling.

**Packet Loss Recovery Module (PLR)**

The IP Telephony solution includes a sophisticated packet loss recovery mechanism, enabling the system to maintain the highest quality of service.

**2.8.2. H.323 Component**

The H.323 Protocol component provides the call signaling and management services as defined in the ITU-T recommendation H.323, *Visual Telephone*

---

* Mean Opinion Score
Systems and Equipment for Local Area Networks Which Provide a Non-Guaranteed Quality of Service.

The H.323 component receives the IP media streams from the IP Network Interface, exchanging the stream contents with resources on the DM3 IPLink board.

Depending on the hardware configuration, the H.323 component can sit either on the host or on the DM3 board. In both cases it is transparent to the host application and communicates via the NetTSC component.
3. Host Application

Dialogic’s IPLink platform provides a standards-based platform for developing Internet telephony applications. It is the user’s responsibility to develop the host application according to the specific needs the application is addressing. Several guidelines are provided in this document, although they are not meant to be specific solutions to specific design problems. Before writing a host application Dialogic Corporation recommends that you thoroughly read this document, the IPTGate Demo Guide and run the IPTGate demo. Many suggestions for application design can be found in the demo.

Topics discussed in this chapter include:
- Host Application Responsibilities
- Host Application Communication with Dialogic Resources

3.1. Host Application Responsibilities

The host application is responsible for all of the connection control, directory services and session control processing, and all of the interaction between the NetTSP and the data network traffic as part of the data transfer. The IPLink platform can be used to build a wide variety of applications, meeting a multitude of user needs.

Following are some typical responsibilities of the host application.
- Overall system management
- PSTN and NetTSP control applications
- Routing

The host application communicates with the IPLink platform is via a high-level API, enabling the application writer to concentrate on application specific issues without having to master the complexities of the IPLink internal design.
3.2. Host Application Communication with Dialogic Resources

The host application communicates with the different Dialogic boards via the Dialogic APIs and other interfaces. Figure 6 presents a sample gateway configuration:

![Gateway Diagram]

**Figure 6. Typical Gateway**

Host application access to DM3 devices on Windows NT is via the DM3 Direct Interface host library functions. The Direct Interface is used in conjunction with the Win32 API to produce highly native applications. The Direct Interface library provides a method for sending and receiving messages, as well as providing a full array of convenience functions for cluster and component management. The DM3 Mediastream Message Block (MMB) is the primary data structure used by an MTNI application to send and receive messages to and from the DM3 platform. See *Using the DM3 Direct Interface for Windows NT* and
3. Host Application

DM3 Direct Interface Function Reference for Windows NT for more information about the Direct Interface.

Host application access to the PSTN Interface Card is via familiar R4 APIs. Refer to the Standard Runtime Library for Windows NT Programmer’s Guide for information about the standard Dialogic Corporation PSTN Card APIs.

3.2.1. Event Reporting

Retrieving events is a key element in designing a WinNT application. In the Windows NT Native R4 environment, the application is notified about an event via the Standard Runtime Library (SRL). In the DM3 environment an application can use any Win32 native synchronization, such as I/O Completion Ports, to wait for event notification.

The Native Synchronization Methods (NSM) solution enables the eventing solution to have one synchronization point in the application for DM3, R4, and other Win32 devices. The NSM solution uses Windows NT native synchronization methods, such as I/O Completion Ports, together with SRL enhancements. Using this approach, DM3 is completely native and R4 devices report their events to this I/O Completion Port using SRL support for NSM.

4. Programming Model

This chapter describes the programming model used by the NetTSP resource. It discusses the general philosophy and method of tracking events. It explains call-states and call-state transitions and how the NetTSC component uses these state-transitions to manage the call.

Topics discussed include:
- What are call-states?
- What are call-state transitions?
- Enabling detection of call-state changes

4.1. What are Call-States?

The NetTSP resource presents a call-control programming model based on the concept of call-state.

A call is a point-to-point multimedia communication between two Internet endpoints. The call begins with the call setup procedure and ends with the call termination procedure. The call consists of the collection of virtual channels between the endpoints.

A call-state is a clearly defined stage of a call’s progression, for example, Initiated, Connected, Disconnected, Idle, etc.

A call is created in one of two ways:
- The host application issues a `TSC_MsgMakeCall` command to initiate an outbound call.
- The H.323 stack notifies the NetTSC component instance that an incoming call is present. This results in a call being created. A call-state transition is generated from the Null state to the Offered state. The host application is notified of the new call via a `Std_MsgEvtDetected` message about this transition to the Offered state and receives the call identifier for this call.
Each call moves from state to state as the call connection is established, connected, and concluded. The particular state a call is in determines what commands and actions may be performed. Such states are called valid states for a particular command. If a command is issued when the call is not in a valid state for that command an error occurs. The following table lists the call-states:

### Table 1. Call-States

<table>
<thead>
<tr>
<th>Call State</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Accepted</td>
<td>The call that was offered has been accepted by the host application. The host application has accepted responsibility in answering the call.</td>
</tr>
<tr>
<td>Alerting</td>
<td>The caller is notified that the call was accepted.</td>
</tr>
<tr>
<td>Connected</td>
<td>The calling and called parties are connected and the call is active on the related call channel. Information may be exchanged. In the case of an outbound call, this state indicates that the remote party has answered. In the case of an inbound call, this state indicates that the local party has answered the call. <strong>This is the only state in which the audio path is established and open.</strong></td>
</tr>
<tr>
<td>Disconnected</td>
<td>The remote party has disconnected from the call.</td>
</tr>
<tr>
<td>Failed</td>
<td>The outbound call attempt was unsuccessful. The call attempt will transition to this state if it is determined to be unsuccessful due to rejection by the other side.</td>
</tr>
<tr>
<td>Idle</td>
<td>The local party to the call has disconnected or has been disconnected.</td>
</tr>
<tr>
<td>Initiated</td>
<td>The outbound call attempt has been initiated (<em>TSC_MsgMakeCall</em>).</td>
</tr>
<tr>
<td>Null</td>
<td>This indicates that the call has been released and all call information related to the call has been destroyed.</td>
</tr>
</tbody>
</table>
4. Programming Model

<table>
<thead>
<tr>
<th>Call State</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Offered</td>
<td>The inbound call is newly arrived and is being offered to the host application. Call information is available to the host application at this time in order for the host application to determine the appropriate action to take with regards to the call.</td>
</tr>
<tr>
<td>Proceeding</td>
<td>Part of the call setup. When the inbound call state transitions to Offered, the outbound state transitions to Proceeding.</td>
</tr>
</tbody>
</table>

4.2. What are Call-State Changes?

Each call maintains a current state, transitioning from its current state to a new state for one of two reasons:

Table 2. State Transitions

<table>
<thead>
<tr>
<th>Reason</th>
<th>Example</th>
</tr>
</thead>
<tbody>
<tr>
<td>The host application requests a transition.</td>
<td>A call that is in the Offered state (an incoming call) will transition to the Connected state if the host application issues a <em>TSC_MsgAnswerCall</em>.</td>
</tr>
<tr>
<td>Changes occur on the network channel associated with the call via the H.323 connection.</td>
<td>A call that is currently in the Connected state will transition to the Disconnected state if the remote party disconnects.</td>
</tr>
</tbody>
</table>

Transition to a new state occurs when a message or event relating to a call is received by the NetTSC component instance. It may be triggered by either the host application or by the network on which the call exists. For example, if the host application sends a *TSC_MsgDropCall* message, the state machine transitions from the Connected state to the Idle state. The host is notified via the *Std_MsgExtDetected* message. See Section 5.2. Detecting Events for more information about events.
4.3. Enabling Detection of Call-State Changes

Call-state changes are reported to the application using the standard DM3 Run Time Control (RTC) mechanism.

Call-state changes (events) must be enabled before they can be reported to the application. The following procedure must be followed for each call-state to be enabled:

1. Send the following message to an appropriate NetTSC instance (see Section 5.3. Allocating a NetTSP for a description of allocating NetTSP instances) for each event to be registered:
   - Std_MsgDetectEvt
   The NetTSC component instance will send a Std_MsgEvtDetected message each time its call state changes to one of those enabled by the application.

2. Repeat step 1 for each event to be enabled.

3. Repeat steps 1 and 2 for each additional NetTSC instance.

NOTE: An alternative to sending Std_MsgDetectEvt for each event is to send Std_MsgDetectxEvts listing all relevant events.

4.4. State Diagrams

The following state diagrams describe the call-states for outbound (to the IP network) and inbound (from the IP network) calls. Each state is represented by an ellipse containing the state name and the Std_MsgEvtDetected message sent to the host upon that state transition. The states are connected with arrows indicating the valid state-changes. Each state change includes the message used to cause that state change.
4. Programming Model

NOTE: These state diagrams are complex. Simplified versions will be shown in Chapter 6 for each call operation.

4.4.1. Outbound Internet Calls

The following state diagram describes the call-states and the state transitions when making an outbound call.
4.4.2. Inbound Internet Calls

The following state diagram describes the call-states and the state transitions when making an inbound call.
4. Programming Model

Figure 9. Inbound Call-State Diagram
5. Using the NetTSP Resource

This chapter explains how the DM3 messages (standard messages, TSC component messages, and NetTSC component messages) are used to implement the various features visible to the host application software. Typical call progressions are presented.

Topics discussed include:

- Messages
- Allocating a NetTSP
- Detecting Events
- Making an Outbound Call
- Receiving an Inbound Call
- Terminating Calls
- Getting Call Statistics
- Other Call Control Operations

5.1. Messages

Communication between the host application and the NetTSC component instances is via messages. This section briefly describes all the messages used by the NetTSC component.

5.1.1. Standard Messages Used by the NetTSC Component

The following DM3 Standard Messages are used by the NetTSC component. Refer to *DM3 Standard Component Interface Messages* for a complete description of all the DM3 Standard Messages.
Table 3. DM3 Standard Messages Used by the NetTSC Component

<table>
<thead>
<tr>
<th>Command Message</th>
<th>Description</th>
<th>Reply Message (indicates success)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Std_MsgCancelAllEvts</td>
<td>Cancels all RTC requests</td>
<td>Std_MsgCancelAllEvtsCmplt</td>
</tr>
<tr>
<td>Std_MsgCancelEvt</td>
<td>Cancel RTC request</td>
<td>Std_MsgCancelEvtCmplt</td>
</tr>
<tr>
<td></td>
<td></td>
<td>MsgCancelEvtCmplt</td>
</tr>
<tr>
<td></td>
<td></td>
<td>MsgCancelEvtCmplt</td>
</tr>
<tr>
<td>Std_MsgCancelxEvts</td>
<td>Cancels multiple RTC requests</td>
<td>Std_MsgCancelxEvtsCmplt</td>
</tr>
<tr>
<td>Std_MsgComtest</td>
<td>Ping message</td>
<td>Std_MsgComtestCmplt</td>
</tr>
<tr>
<td>Std_MsgDetectEvt</td>
<td>Detect RTC event</td>
<td>Std_MsgDetectEvtCmplt</td>
</tr>
<tr>
<td>Std_MsgDetectxEvts</td>
<td>Detect multiple RTC events</td>
<td>Std_MsgDetectxEvtsCmplt</td>
</tr>
<tr>
<td>Std_MsgError</td>
<td>Error detected</td>
<td>Not applicable</td>
</tr>
<tr>
<td>Std_MsgEvtDetected</td>
<td>Event detected</td>
<td>Not applicable</td>
</tr>
<tr>
<td>Std_MsgExit</td>
<td>Shuts down an instance in a proprietary way.</td>
<td>Std_MsgExitCmplt</td>
</tr>
<tr>
<td>Std_MsgGetParm</td>
<td>Read parameter value</td>
<td>Std_MsgGetParmCmplt</td>
</tr>
<tr>
<td>Std_MsgInit</td>
<td>Initialize resource</td>
<td>Std_MsgInitCmplt</td>
</tr>
</tbody>
</table>
5. Using the NetTSP Resource

<table>
<thead>
<tr>
<th>Command Message</th>
<th>Description</th>
<th>Reply Message</th>
</tr>
</thead>
<tbody>
<tr>
<td>Std_MsgSetAllParmsDef</td>
<td>Sets parameter values</td>
<td>Std_MsgSetAllParmsDefCmplt</td>
</tr>
<tr>
<td></td>
<td>to default values</td>
<td></td>
</tr>
<tr>
<td>Std_MsgSetParm</td>
<td>Set parameter</td>
<td>Std_MsgSetParmCmplt</td>
</tr>
<tr>
<td>Std_MsgSetParmDef</td>
<td>Changes the value of a</td>
<td>Std_MsgSetParmDefCmplt</td>
</tr>
<tr>
<td></td>
<td>parameter</td>
<td></td>
</tr>
</tbody>
</table>

5.1.2. TSC Messages Used by the NetTSC Component

The following TSC Messages are used by the NetTSC component. Refer to the IPT Reference Guide for a complete description of all the TSC Messages.

Table 4. TSC Component Messages Used by the NetTSC Component

<table>
<thead>
<tr>
<th>Command Message</th>
<th>Description</th>
<th>Reply Message</th>
</tr>
</thead>
<tbody>
<tr>
<td>TSC_MsgAcceptCall</td>
<td>Accepts the incoming call</td>
<td>Std_MsgEvtDetected(Accepted)</td>
</tr>
<tr>
<td>TSC_MsgAnswerCall</td>
<td>Answers the incoming call</td>
<td>Std_MsgEvtDetected(Connected)</td>
</tr>
<tr>
<td>TSC_MsgDropCall</td>
<td>Drops or disconnects a call</td>
<td>Std_MsgEvtDetected(Idle)</td>
</tr>
</tbody>
</table>
### Command Message and Reply Message Table

<table>
<thead>
<tr>
<th>Command Message</th>
<th>Description</th>
<th>Reply Message</th>
</tr>
</thead>
<tbody>
<tr>
<td>TSC_MsgGetCallInfo</td>
<td>Requests stored information related to a call</td>
<td>TSC_MsgGetCallInfoCmplt</td>
</tr>
<tr>
<td>TSC_MsgGetCallState</td>
<td>Requests the current state of a call</td>
<td>TSC_MsgGetCallStateCmplt</td>
</tr>
<tr>
<td>TSC_MsgMakeCall</td>
<td>Initiates a call to a specified destination address</td>
<td>TSC_MsgMakeCallCmplt</td>
</tr>
<tr>
<td></td>
<td>DestAddr is the IP address, rather than the phone number.</td>
<td></td>
</tr>
<tr>
<td>TSC_MsgReleaseCall</td>
<td>Releases a call identifier</td>
<td>Std_MsgEvtDetected(Null)</td>
</tr>
<tr>
<td>TSC_MsgRejectCall</td>
<td>Rejects an incoming call</td>
<td>Std_MsgEvtDetected(Idle)</td>
</tr>
</tbody>
</table>

### 5.1.3. NetTSC Component Messages

The following messages are specific to the NetTSC component.
5. Using the NetTSP Resource

Table 5. NetTSC Component Messages

<table>
<thead>
<tr>
<th>Command Message</th>
<th>Description</th>
<th>Reply Message (indicates success)</th>
</tr>
</thead>
<tbody>
<tr>
<td>NetTSC_MsgSendUserInputIndication</td>
<td>Puts DTMF input on the H.245 channel</td>
<td>NA</td>
</tr>
<tr>
<td>NetTSC_MsgSendNonStandardCmd</td>
<td>Sends non-standard data on the H.245 channel, between two Dialogic gateways</td>
<td>NA</td>
</tr>
</tbody>
</table>

5.2. Detecting Events

This section describes detecting events in detail.

5.2.1. About Events

An event is an asynchronous message sent by a resource component to inform the host application of a change in state or other information.
5.2.2. Event Messages

The components and component instances of the NetTSC component use two standard messages to communicate information about events:

<table>
<thead>
<tr>
<th>Standard Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Std_MsgDetectEvt</td>
<td>Performs two functions:</td>
</tr>
<tr>
<td></td>
<td>• enables a component or component instance to detect the event specified by the message</td>
</tr>
<tr>
<td></td>
<td>• registers the host application to receive notification whenever the specified event occurs.</td>
</tr>
<tr>
<td>Std_MsgEventDetected</td>
<td>Alerts the host application registered for notification about this particular event.</td>
</tr>
</tbody>
</table>

The *Std_MsgEventDetected* message is the only message that conveys event information. To accommodate a variety of events, the *Std_MsgEventDetected* message has a variable body; the type of event that has occurred determines the type and number of fields inside the message.

Each DM3 component has its own set of valid event types. NetTSC component event types are discussed in Section 5.2.4. *Events Specific to the NetTSC Component*.

5.2.3. The Event-Reporting Process

In order for event reporting to occur, all these steps must be performed in the specified order:
5. Using the NetTSP Resource

1. The host application interested in a particular event must register for notification with the component or component instance that will encounter the event. Send a **Std_MsgDetectEvt** message to the appropriate NetTSC component instance providing: the event type, e.g., **TSCEvtCallState_Offered**; an appropriate Event Label (generated by the application); and the address to send the event to (i.e., the application’s address) as the body of the message.

2. The specified event occurs.

3. The component instance that encountered the event sends a **Std_MsgEvtDetected** event message to the host application. The type of event determines the body and fields of this message.

5.2.4. Events Specific to the NetTSC Component

Table 6 shows the events specific to the NetTSC component. Depending on the event that has occurred, the body of the **Std_MsgEvtDetected** message will contain specific fields related to that event. The shaded rows indicate that the event is not supported in the current revision of the IPLink development kit.

**Table 6. NetTSC Component Event Types**

<table>
<thead>
<tr>
<th>Event</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>TSCEvtCallInfo</td>
<td>Delivers call-related information to the application as the information becomes available.</td>
</tr>
<tr>
<td>TSCEvtCallState</td>
<td>Indicates a change in call state.</td>
</tr>
<tr>
<td>NetTSCEvtH245Data_NonStdCmd</td>
<td>Notifies the application that non-standard data has been received on the H.245 channel.</td>
</tr>
</tbody>
</table>
### Event Description

<table>
<thead>
<tr>
<th>Event</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>NetTSC_EvtH245Data_UsrInputIndication</td>
<td>Notifies the application that the User-Input-Indication data has been received on the H.245 channel.</td>
</tr>
<tr>
<td>NetTSC_EvtThresholdAlarm</td>
<td>Notifies that an alarm occurs.</td>
</tr>
<tr>
<td>NetTSC_EvtSystemFailed</td>
<td>H.323 failure, e.g., NIC disconnect</td>
</tr>
</tbody>
</table>

### 5.3. Allocating a NetTSP

Before a call can be opened, a cluster must be allocated. This identifies, for the host application, which cluster is to manage the call.

#### 5.3.1. Procedure: Allocating a Cluster

The host application allocates a cluster by issuing a `mntClusterAllocate` message followed by an `mntClusterCombyAttribute` message.

The steps involved are as follows:

**Table 7. Procedure: Allocating a Cluster**

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host allocates a NetTSP cluster</td>
<td><code>mntClusterAllocate</code> (sent from application to Direct Interface)</td>
<td>Selects an H.323 cluster</td>
</tr>
</tbody>
</table>
5. Using the NetTSP Resource

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>2</td>
<td>Host identifies the NetTSC component</td>
<td><em>mntClusterComphyAttribute</em> (sent from application to Direct Interface)</td>
<td>Identifies the NetTSC component within the cluster.</td>
</tr>
</tbody>
</table>

5.4. Assigning Timeslots

In order for data to be transmitted to and received from the SCbus, timeslots must be assigned to the cluster’s components. A transmit timeslot is automatically assigned to the NetTSC component during the download procedure. The receive timeslot must be assigned by the application. The procedure for assigning a receive timeslot to the NetTSC component is described here.

5.4.1. Procedure: Assigning a Receive Timeslot to the NetTSC Component

The host application assigns a receive timeslot by issuing a *mntClusterTSAssign* message followed by a *mntClusterActivate* message.

The steps involved are as follows:
### Table 8. Procedure: Assigning a Timeslot

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host assigns a receive timeslot to a NetTSC instance</td>
<td><em>xx_mntClusterTSAssign</em> (sent from application to Direct Interface)</td>
<td>Assigns a receive timeslot to a NetTSC component instance.</td>
</tr>
<tr>
<td>2</td>
<td>Host activates the connection</td>
<td><em>mntClusterActivate</em> (sent from application to Direct Interface)</td>
<td>Activates the IN-port in the NetTSC component instance. This allows TDM data to flow from the SCbus into the NetTSC component instance.</td>
</tr>
</tbody>
</table>

### 5.4.2. Querying About the SCbus Transmit Timeslot

Components can receive data only when they are aware of the transmitting component’s Tx timeslot. Each side must query the other to determine the transmit timeslot.

**Listening to the PSTN Card**

The NetTSC component must query the PSTN card to determine the transmit slot. The procedure is as follows:
5. Using the NetTSP Resource

Table 9. Procedure: Querying the PSTN

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Get the SCbus transmit timeslot assigned to the PSTN card</td>
<td>$xx_{getXmitSlo}$ (where $xx$ indicates the type of PSTN card)</td>
<td>The legacy card Tx slot is identified</td>
</tr>
<tr>
<td>2</td>
<td>Assign the timeslot to the NetTSP cluster’s Rx port</td>
<td>$mntClusterTSAssign$</td>
<td>Assigns a receive timeslot to the NetTSC component instance.</td>
</tr>
<tr>
<td>3</td>
<td>Activate the connection</td>
<td>$mntClusterActivate$</td>
<td>Activates the IN-port in the NetTSP cluster. This allows TDM data to flow from the SCbus into the NetTSP cluster.</td>
</tr>
</tbody>
</table>

**Listening to the DM3 Card**

The legacy card must query the NetTSP cluster to determine the Transmit slot. The procedure is as follows:
Table 10. Procedure: Querying the DM3

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Get the SCbus transmit timeslot assigned to the NetTSP cluster</td>
<td>mntClusterSlotInfo</td>
<td>The NetTSP cluster Tx slot is identified</td>
</tr>
<tr>
<td>2</td>
<td>Assign the timeslot to the legacy card Rx port</td>
<td>xx_listen (where xx indicates the type of PSTN card)</td>
<td>Assigns a receive timeslot to the legacy card.</td>
</tr>
</tbody>
</table>

5.5. Making an Outbound Call

This section describes a typical procedure for making an outbound call to the IP Network. A simplified call session model is presented here, illustrating the call-state transitions: Null → Initiated → Connected → Idle → Null. Other states will be added in later sections.

**NOTE:** This section, and the sections that follow, present only one possible way of implementing these call control scenarios. It is the responsibility of the application developer to select the implementation most suitable for the particular application needs.

The state diagram below illustrates the state transitions for making an outbound call. See Section 5.7, Terminating Calls for an explanation of the call termination options.
5. Using the NetTSP Resource

5.5.1. Procedure: Making an Outbound Call

The host application places an outgoing call to the IP Network by issuing a
TSC_MsgMakeCall message. The Call State RTC mechanism is used to monitor
the outcome of the call.

The steps involved are as follows:
### Table 11. Procedure: Outbound Calls

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host registers events</td>
<td><code>Std_MsgDetectxEvts (event_list)</code> (sent from application to NetTSC component instance)</td>
<td>All relevant events are registered with the kernel.</td>
<td>Null</td>
</tr>
<tr>
<td>2</td>
<td>Host initiates call</td>
<td><code>TSC_MsgMakeCall</code> (sent from application to NetTSC component instance)</td>
<td>The NetTSC component instance returns <code>TSC_MsgMakeCallCmplt</code> with the call ID. Logical channels are opened and data streams are allocated. Note that a <code>Std_MsgError</code> message could be received instead, indicating that an error occurred trying to initiate the outgoing call.</td>
<td>Initiated</td>
</tr>
<tr>
<td>3</td>
<td>Host is notified of connection</td>
<td><code>Std_MsgEvtDetected (Connected)</code> (sent from NetTSC component instance to application)</td>
<td>Call-state transitions to Connected.</td>
<td>Connected</td>
</tr>
</tbody>
</table>
5. Using the NetTSP Resource

NOTE: The host application does not have to register all events. For example, in this call session the application might not want to be notified of a transition to the Initiated state. The application should then register only the Null, Connected, Disconnected, and Idle states. The state-transition to Initiated will occur without notifying the application. When the call-state transitions to Connected, a *Std_MsgEvtDetected(Connected)* message is sent to the host application.

5.6. Receiving an Inbound Call

This section describes the procedure for receiving an inbound call from the IP Network. A simplified call session model is presented first, illustrating the call-state transitions: Null → Offered → Connected → Idle → Null.

When the host application receives the event message *Std_MsgEvtDetected(Offered)*, indicating that an incoming call is being offered, it has a choice of operations on the call:

- **answering** the call by issuing a *TSC_MsgAnswerCall* message
  
  This command instructs the NetTSC component instance to open the audio streams and allow conversation to begin. The call-state transitions to Connected.

- **accepting** a call by issuing a *TSC_MsgAcceptCall* message

  This command instructs the NetTSC component instance to respond to the calling party while waiting to open the audio streams. The call-state transitions to an interim state – Accepted. This keeps the calling party from “timing-out” while the gateway performs some other action. A *TSC_MsgAnswerCall* message must be sent to open the audio channels and enable conversation.
• **rejecting** the call by issuing a *TSC_MsgRejectCall* message

  This command instructs the NetTSC component instance to reject the call. The call-state transitions to Idle. Call information is still available (e.g., calling party) until the *TSC_MsgReleaseCall* message is sent. See Section 5.9.1, *Rejecting an Inbound Call* for a description of rejecting a call.

### 5.6.1. Procedure: Answering an Inbound Call

The state diagram below illustrates the state transitions for answering an inbound call. See Section 5.7, *Terminating Calls* for an explanation of the call termination options.
Figure 11. Answer Inbound Call - Simplified State Diagram

The host application receives incoming call notification via the Standard DM3 Event mechanism.

The steps involved are as follows:
Table 12. Procedure: Answering Inbound Calls

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host registers events</td>
<td>Std_MsgDetectxEvts (event_list)</td>
<td>All relevant events are registered with the kernel.</td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host notified of incoming call</td>
<td>Std_MsgEvtDetected (Offered, CallID)</td>
<td>Call-state changed from Idle to Offered</td>
<td>Offered</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from NetTSC component to application)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>Host answers call</td>
<td>TSC_MsgAnswerCall (CallID, coder)</td>
<td>Logical channels are opened and data streams are allocated.</td>
<td>Connected</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>4</td>
<td>Host is notified of connection</td>
<td>Std_MsgEvtDetected (Connected)</td>
<td>Call-state transitions to Connected</td>
<td>Connected</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from NetTSC component to application)</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
Message Summary

The following figure illustrates the messages and call-states on either side of an Internet gateway for the above model.

![Call Establishment Messages - Simple Model](image)

Figure 12. Call Establishment Messages - Simple Model

5.6.2. Procedure: Accepting a Call

A gateway may accept a call without opening the audio path. This allows the gateway to respond to the call offering, attempt to contact the destination address, and only after the destination answers the call, to open the audio path.
This feature allows the application developer to begin the billing procedure only when a call is actually connected.

The state diagram below illustrates the state transitions for the above inbound call. See Section 5.7. Terminating Calls for an explanation of the call termination options.

![State Diagram](image)

**Figure 13. Accept Inbound Call - Simplified State Diagram**
The host application receives incoming call notification via the Standard DM3 Event mechanism.

The steps involved are as follows:

**Table 13. Procedure: Accepting Inbound Calls**

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host registers events</td>
<td>Std_MsgDetectxEvts(event_list)</td>
<td>All relevant events are registered with the kernel.</td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host notified of incoming call</td>
<td>Std_MsgEvtDetected(Offered, CallID)</td>
<td>Call-state changed from Idle to Offered</td>
<td>Offered</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from NetTSC component to application)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>Host accepts call</td>
<td>TSC_MsgAcceptCall(CallID)</td>
<td>The inbound call is acknowledged, but not yet answered.</td>
<td>Accepted</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td>This interim state allows the gateway to perform other actions before connecting the call.</td>
<td></td>
</tr>
</tbody>
</table>
### Step Operation Message Description State

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>4</td>
<td>Host answers call</td>
<td><strong>TSC_MsgAnswerCall</strong> <em>(CallID, coder)</em></td>
<td>Logical channels are opened and data streams are allocated.</td>
<td>Connected</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td><strong>NOTE:</strong> The audio path is opened only when <strong>TSC_MsgAnswerCall</strong> is sent.</td>
<td></td>
</tr>
<tr>
<td>5</td>
<td>Host is notified of connection</td>
<td><strong>Std_MsgEvtDetected</strong> <em>(Connected)</em></td>
<td>Call-state transitions to Connected</td>
<td>Connected</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from NetTSC component to application)</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

### Message Summary

The following figure illustrates the messages and call-states on either side of an Internet gateway for the above model.
5. Using the NetTSP Resource

5.7. Terminating Calls

Once a call has been established (call-state = Connected), it can be terminated by either the host application (using call control operations) or the network (using H.323 Protocol termination).

The state diagram below illustrates the state transitions for terminating a call:
5.7.1. Procedure: Terminating a Call by the Host

The host application can terminate a call using the command `TSC_MsgDropCall`. This will result in two actions:

- a call state transition to Idle
- `Std_MsgEvtDetected(Idle)` is sent to the host application, if it is registered for this type of event notification.

**NOTE:** Call statistics are stored until the call state transitions to Null. See Section 5.8, *Getting Call Statistics* for the procedure for accessing the call statistics.
5. Using the NetTSP Resource

The steps involved are as follows:

**Table 14. Terminating a Call by the Host**

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host application disconnects</td>
<td>TSC_MsgDropCall</td>
<td>Tells the NetTSC instance to begin the sequence to disconnect the call.</td>
<td>Idle</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application</td>
<td>The NetTSC instance returns Std_MsgEvtDetected (Idle) to the application upon successful completion of the disconnect.</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>to NetTSC component</td>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host clears the call</td>
<td>TSC_MsgReleaseCall</td>
<td>CallID is cleared and the call state transitions to Idle.</td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application</td>
<td>The NetTSC instance returns Std_MsgEvtDetected (Null) to the application upon successful completion of the disconnect.</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>to NetTSC component</td>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>instance)</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

5.7.2. Procedure: Call Terminated by the IP Network

The network can terminate the call while the call is in the Connected state. This will result in two actions:

- a call state transition to Disconnected

- Std_MsgEvtDetected(Disconnected) is issued to the application to report the transition, if the application is registered for this type of event notification.
The steps involved are as follows:

**Table 15. Call Terminated by the IP Network**

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Network disconnects</td>
<td><strong>TSC_MsgDisconnected</strong>&lt;br&gt;(sent from H.323 component to NetTSC component instance)</td>
<td>Tells the NetTSC instance that the call has been disconnected.&lt;br&gt;The NetTSC instance returns <strong>Std_MsgEvtDetected (Disconnected)</strong> to the application.</td>
<td>Disconnected</td>
</tr>
<tr>
<td>2</td>
<td>Host application disconnects</td>
<td><strong>TSC_MsgDropCall</strong>&lt;br&gt;(sent from application to NetTSC component instance)</td>
<td>Tells the NetTSC instance to begin the sequence to disconnect the call.&lt;br&gt;The NetTSC instance returns <strong>Std_MsgEvtDetected (Idle)</strong> to the application upon successful completion of the disconnect.</td>
<td>Idle</td>
</tr>
</tbody>
</table>
5. Using the NetTSP Resource

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>3</td>
<td>Host clears the call</td>
<td>TSC_MsgReleaseCall</td>
<td>CallID is cleared and the call state transitions to Idle. The NetTSC instance returns Std_MsgEvtDetected (Null)</td>
<td>Null</td>
</tr>
</tbody>
</table>

**Message Summary**

The following figure illustrates the messages and call-states on either side of an Internet gateway for the above model.
5.8. Getting Call Statistics

The host application can get call statistics from the NetTSC component instance by issuing either a `TSC_MsgGetCallInfo` or a `TSC_MsgGetCallInfoExt` message. Information can be gathered at any point during a call, until the call state transitions to Null.
5. Using the NetTSP Resource

5.8.1. Procedure: Getting Call Statistics

The following procedure illustrates how to get call statistics at the end of a call.

**Table 16. Getting Call Statistics**

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host application</td>
<td><em>TSC_MsgDropCall</em></td>
<td>Tells the NetTSC instance to begin the sequence to disconnect the call.</td>
<td>Idle</td>
</tr>
<tr>
<td></td>
<td>disconnects</td>
<td>(sent from application to NetTSC component instance)</td>
<td>The NetTSC instance returns <em>Std_MsgEvtDetected (Idle)</em> to the application upon successful completion of the disconnect.</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host requests call</td>
<td><em>TSC_MsgGetCallInfo</em></td>
<td>The host application request call information from the NetTSC instance.</td>
<td>NA</td>
</tr>
<tr>
<td></td>
<td>information</td>
<td>(sent from application to NetTSC component instance)</td>
<td>The NetTSC instance returns <em>TSC_MsgGetCallInfoCmplt</em> with the requested information.</td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>Host clears the call</td>
<td><em>TSC_MsgReleaseCall</em></td>
<td>CallID is cleared and the call state transitions to Idle.</td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td>The NetTSC instance returns <em>Std_MsgEvtDetected (Null)</em></td>
<td></td>
</tr>
</tbody>
</table>
5.9. Other Call Control Operations

5.9.1. Rejecting an Inbound Call

This section describes the procedure for rejecting a call.

The state diagram below illustrates the state transitions for rejecting a call:

![State Diagram](image)

Figure 17. Rejecting an Inbound Call - Simplified State Diagram
The following procedure illustrates how to reject an inbound call:

### Table 17. Rejecting an Inbound Call

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host notified of incoming call</td>
<td><em>Std_MsgEvtDetected (Offered, CallID)</em></td>
<td>Call-state changed from Idle to Offered</td>
<td>Offered</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from NetTSC component to application)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host rejects call</td>
<td><em>TSC_MsgRejectCall</em></td>
<td>The inbound call is rejected.</td>
<td>Idle</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>Host releases call</td>
<td><em>TSC_MsgReleaseCall</em></td>
<td>Call ID is cleared and the call state transitions to Idle. The NetTSC instance returns <em>Std_MsgEvtDetected (Null)</em></td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

### 5.9.2. Failed Outbound Call

This section describes what happens when an outbound call fails.

The state diagram below illustrates the state transitions when a call is rejected by the IP Network:
Figure 18. Failed Outbound Call - Simplified State Diagram

The following procedure illustrates how an outbound call fails, i.e., is rejected by the IP network:
5. Using the NetTSP Resource

Table 18. Failed Outbound Call

<table>
<thead>
<tr>
<th>Step</th>
<th>Operation</th>
<th>Message</th>
<th>Description</th>
<th>State</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Host registers events</td>
<td>Std_MsgDetectxEvts (event_list)</td>
<td>All relevant events are registered with the kernel.</td>
<td>Null</td>
</tr>
<tr>
<td></td>
<td></td>
<td>(sent from application to NetTSC component instance)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Host initiates call</td>
<td>TSC_MsgMakeCall</td>
<td>The NetTSC component instance returns TSC_MsgMakeCallCmplt with the call ID. Logical channels are opened and data streams are allocated. Note that a Std_MsgError message could be received instead, indicating that an error occurred trying to initiate the outgoing call.</td>
<td>Initiated</td>
</tr>
</tbody>
</table>
### Step 3: Call is rejected by the IP network
- **Message:** `MsgFailed`
  - (sent from the H.323 component to the NetTSC component instance)
  - Description: The H.323 component receives a rejection from the IP Network and informs the NetTSC component instance. The NetTSC component instance returns `Std_MsgEvtDetected (Failed)` to the host application.
  - State: Failed

### Step 4: Host application disconnects
- **Message:** `TSC_MsgDropCall`
  - (sent from the host application to the NetTSC component instance)
  - Description: Tells the NetTSC instance to begin the sequence to disconnect the call. The NetTSC instance returns `Std_MsgEvtDetected (Idle)` to the application upon successful completion of the disconnect.
  - State: Idle

### Step 5: Host releases call
- **Message:** `TSC_MsgReleaseCall`
  - (sent from application to NetTSC component instance)
  - Description: Call ID is cleared and the call state transitions to Idle. The NetTSC instance returns `Std_MsgEvtDetected (Null)`
  - State: Null
5. Using the NetTSP Resource
6. Setting IPLink Parameters

Various parameters may be set to configure the IPLink platform for an application’s particular needs. This chapter discusses the content of each parameter and includes examples of how they are set in the parameter configuration files supplied by Dialogic Corporation.

Topics discussed include:
- SP parameters
- H.323 parameters
- Miscellaneous parameters
- Setting number of instances
- H.323 Stack configuration

6.1. FCD File

The Feature Configuration Description (FCD) file defines certain components that need to be configured with a unique set of parameters. The parameters are defined in the DM3 messages and sent to the components.

A default FCD file, ipt.fcd, is supplied with the IPLink platform software. The following sections describe the IPLink parameters and how to set them. There are three types of IPLink parameters set in the FCD file:
- SP parameters
- H.323 parameters
- Miscellaneous parameters

The relevant sections of the ipt.pcd file can be found in Appendix C.
6.1.1. SP Parameters

Echo Canceler

Two parameters can be set for the echo canceler:

- Enable/Disable

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>prmECActive</td>
<td>0x1b12</td>
</tr>
</tbody>
</table>

Possible values are:

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>Enable (Default)</td>
</tr>
<tr>
<td>1</td>
<td>Disable</td>
</tr>
</tbody>
</table>

- Number of taps (1 tap = 1/8 msec.)

The echo canceler can be set to sample up to 256 taps (32 msecs).

**NOTE:** While the echo canceler is designed to support an “echo-tail” of up to 64 msecs, due to testing limitations this parameter has been limited to a maximum value of 32 msecs.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>prmECOrder</td>
<td>0x1b13</td>
</tr>
</tbody>
</table>

Possible values are:

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Any integer between: 32 – 256 (Default = 128)</td>
</tr>
</tbody>
</table>
6. Setting IPLink Parameters

**Power Level**

The power level parameter attenuates the power level of the encoded frames. It is computed by multiplying the encoded frame by the value of the PwrLvlCtrl parameter:

\[ \text{PowerLevel} = \text{EncodedFrame} \times \text{PwrLvlCtrl parameter} \]

Each sample of decoded PCM data is multiplied by this factor before being transmitted to the bus.

The default is 0.5.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>prmPwrLvlCtrl</td>
<td>0x1b14</td>
</tr>
</tbody>
</table>

Possible values are:

| Any number between: | 0 – 0.9999999 (0x0 – 0x7fffff) |
| Default value:      | 0.5 (0x400000)                 |

**NOTE:** The parameter must assume a hexadecimal value corresponding to the DSP’s 24 bit fractional representation.

To compute the value:

1. Choose the desired attenuation factor in the range 0.0 – 0.999999.
2. Multiply that value by \(2^{23}\).
3. Convert that value to its hexadecimal representation and set as the VolumeControl parameter.

**Automatic Gain Control (AGC)**

The AGC maintains a uniform signal power level as the data is retrieved from the bus.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>prmAGCActive</td>
<td>0x1b1c</td>
</tr>
</tbody>
</table>

Possible values are:

<table>
<thead>
<tr>
<th>0</th>
<th>Enable (Default)</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Disable</td>
</tr>
</tbody>
</table>

**High Pass Filter (HPF)**

The HPF removes DC and very low frequency corruption of the data.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>prmHPFActive</td>
<td>0x1b1d</td>
</tr>
</tbody>
</table>

Possible values are:

| 0 | Enable (Default) |
6. Setting IPLink Parameters

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmDialogicEnable</td>
<td>0x1e19</td>
</tr>
</tbody>
</table>

6.1.2. H.323 Parameters

**Dialogic Enable**

If set, enables Dialogic specific features. This parameter must be set to use the `TSC_MsgNonStandardCmd` message.
Possible values are:

<p>| | |</p>
<table>
<thead>
<tr>
<th></th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>Standard Gateway (Default)</td>
</tr>
<tr>
<td>1</td>
<td>Dialogic Gateway</td>
</tr>
</tbody>
</table>

**Debug Print Levels**

Sets the debug print level for the:

- Stack module (PrmDebugLevelStack)
- Message module (PrmDebugLevelMsg)
- Stream module (PrmDebugLevelStream)
- State Machine module (PrmDebugLevelStates)
- Timer module (PrmDebugLevelTimer)
- Utilities module (PrmDebugLevelUtil)
- MNTI module (PrmDebugLevelMNTI)
- RV module (PrmDebugLevelRVSTACK)

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmDebugLevelStack</td>
<td>0x1e0e</td>
<td>2</td>
</tr>
<tr>
<td>PrmDebugLevelMsg</td>
<td>0x1e0f</td>
<td>3</td>
</tr>
<tr>
<td>PrmDebugLevelStream</td>
<td>0x1e10</td>
<td>0</td>
</tr>
<tr>
<td>PrmDebugLevelStates</td>
<td>0x1e11</td>
<td>3</td>
</tr>
<tr>
<td>PrmDebugLevelTimer</td>
<td>0x1e12</td>
<td>0</td>
</tr>
<tr>
<td>PrmDebugLevelUtil</td>
<td>0x1e13</td>
<td>3</td>
</tr>
</tbody>
</table>
### 6. Setting IPLink Parameters

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmDebugLevelMNTI</td>
<td>0x1e14</td>
<td>0</td>
</tr>
<tr>
<td>PrmDebugLevelRVSTACK</td>
<td>0x1e1e</td>
<td>0</td>
</tr>
</tbody>
</table>

Possible values are:

<table>
<thead>
<tr>
<th>Value</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>no printouts</td>
</tr>
<tr>
<td>1</td>
<td>fatal errors only</td>
</tr>
<tr>
<td>2</td>
<td>adds non-fatal error printouts</td>
</tr>
<tr>
<td>3</td>
<td>adds warning printouts</td>
</tr>
<tr>
<td>4</td>
<td>adds trace printouts</td>
</tr>
</tbody>
</table>

### 6.1.3. Miscellaneous Parameters

#### Compute Call Duration

This parameter specifies the method used to compute the call duration:

- When the NetTSC transitions to Connected state
- When the NetTSC transitions to Initiated state

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmCallDurationComput</td>
<td>0x1e1a</td>
</tr>
</tbody>
</table>
Possible values are:

<table>
<thead>
<tr>
<th>Parameter Value</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>0 (default)</td>
<td>Compute call duration from the Connected state</td>
</tr>
<tr>
<td>1</td>
<td>Compute call duration from the Initiated state</td>
</tr>
</tbody>
</table>

### Set Number of Read Stream Buffers

This parameter sets the number of packets buffered between the VSR component and the H.323 component. If there is excessive load on the host, or many interruptions on the IP network, you may need to set this parameter to a higher value. Otherwise, Dialogic recommends leaving it at its default value.

**NOTE:** This parameter is used only with DM/IPLink-T1 or DM/IPLink-E1 boards.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmNumOfReadStreamBuf</td>
<td>0x1e15</td>
</tr>
</tbody>
</table>
Possible values are:

| Any integer between: | 1 - 10 (Default = 1) |

**Set Number of Write Stream Buffers**

This parameter sets the number of packets buffered between the H.323 component and the VSR component. If there is excessive load on the host, or many interruptions on the IP network, you may need to set this parameter to a higher value. Otherwise, Dialogic recommends leaving it at its default value.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmNumOfWriteStreamBuf</td>
<td>0x1e16</td>
</tr>
</tbody>
</table>
Possible values are:

| Any integer between: | 1 - 40 (Default = 1) |

**Setting the Number of Write DTMF Buffers**

This parameter sets the number of out-of-band DTMF packets buffered between the VSR component and the H.323 component.

<table>
<thead>
<tr>
<th>Parameter Name</th>
<th>Parameter Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>PrmNumOfWriteDTMFBuf</td>
<td>0x1e22</td>
</tr>
</tbody>
</table>
6. Setting IPLink Parameters

Possible values are:

| Any integer between: | 1 - 40 (Default = 10) |

6.2. PCD File

The Product Configuration Description (PCD) file describes the configurable parameters for one or more Processor Load Modules (PLMs) that are defined for a single platform or board. You can change the number of instances, depending on your IP telephony application configuration.

**CAUTION**

Do not change any settings in the PCD file, except for those explained here. The firmware will not function properly if other changes are made.

Use the following procedure to change the PCD file:

1. Locate the [CP] section
2. Change the **NumInstances** field for the following sections:
   - [COMP VSRP]
   - [COMP H323I]
   - [COMP NetTSC]
   - [COMP HOST]
   - [SP ONYX <1–6>] (change to <1–n>, where n = number of instances)
For example, locate the [COMP VSRP] section. Change the **NumInstances** field from 6 channels to 4 channels.

```
[COMP VSRP]
{
  Attribute : Std_ComponentType:0x1a
  NumInstances : 6
  StartInstanceNum : 1
  ConfigOption : YES
  InitOption : YES
  DependentComp : CP TSC
}
```

### 6.3. Config.val

You can configure the H.323 Stack Configuration file (config.val) to:

- Describe coder capabilities
- Enable a gatekeeper

These capabilities are broadcast during the capability exchange at call setup.

The entire config.val file can be found in Appendix B.

---

**CAUTION**

Do not change any settings in the config.val file, except for those explained here. The H.323 stack will not function properly if other changes are made.
6. Setting IPLink Parameters

6.3.1. Describe Coder Capabilities

Each supported coder’s capabilities are described in the `config.val` file. G.711 and G.723.1 are currently designated as the default coders. You may adjust certain receive parameters for each coder.

**G.711**

You may set the maximum frame size that this coder can receive. It can be set for either 30, 20, or 10 msecs. The larger frame size also allows the smaller frame sizes, i.e., the default 30 msec frame size also allows the 20 msec and 10 msec frame sizes. The line to be changed is highlighted in the code fragment below:

```plaintext
4    capabilityTable = 0
5     * = 0 #Sequence
6 capabilityTableEntryNumber = 1 # INTEGER [1..65535]
6     capability = 0
7     receiveAudioCapability = 0
8    g711Ulaw64k = 30 # INTEGER [1..256]
```

**G.723.1**

You may designate the number of frames per packet that the coder can receive, and enable or disable silence suppression (VAD). As of the time of the writing of this guide, the IPLink platform can receive up to 8 frames per packet. See the Release Notes for any changes. The lines to be changed are highlighted in the code fragment below:

```plaintext
5     * = 0 #Sequence
6 capabilityTableEntryNumber = 2 # INTEGER [1..65535]
6     capability = 0
7     receiveAudioCapability = 0
8    g7231 = 0
```
NOTE: The capabilityTableEntryNumber must be sequential starting from 1. If you make any changes, make sure that the numbers are correct. In addition, the capabilityDescriptors section must contain the same number of capabilities. Comment out non-applicable lines:

4  capabilityDescriptors = 0
5   *= 0 #Sequence
6  capabilityDescriptorNumber = 0 #INTEGER [1..255]
7    *= 0
8     *= 1 # INTEGER [1..65535]
8     *= 2 # INTEGER [1..65535]
#8    *= 3 # INTEGER [1..65535]
#8    *= 4 # INTEGER [1..65535]

6.3.2. Enable a Gatekeeper

A gatekeeper serves as a directory of Internet addresses and telephone numbers. The IPLink platform can be configured to register with a gatekeeper. By default, the IPLink platform is configured for no gatekeeper as shown in the following fragment from the config.val file:

1  RAS = 0
2  responseTimeOut = 1
#2  gatekeeper = 1
2  manualRAS = 1
#2  manual Discovery = 0
#3  defaultGatekeeper = 0
#4  ipAddress = 0
#5   ip = <200.202.200.200>
#5   port = 1719

Enter an integer from 1 to 8
0 = Off
1 = On
6. Setting IPLink Parameters

To register with a gatekeeper, comment line 4 (of this section of code). It should appear as follows:

```plaintext
1 RAS = 0
2 responseTimeOut = 1
#2 gatekeeper = 1
#2 manualRAS = 1
#2 manual Discovery = 0
#3 defaultGatekeeper = 0
#4 IPAddress = 0
#5 ip = <200.202.200.200>
#5 port = 1719
```

In addition, you must either insert a specific gatekeeper IP address, or enable discovery (broadcasting a request for a gatekeeper).

If you know the IP address of the gatekeeper you want to register with, uncomment lines 5 – 9 and insert the correct IP address in line 8. It should appear as follows:

```plaintext
1 RAS = 0
2 responseTimeOut = 1
#2 gatekeeper = 1
#2 manualRAS = 1
2 manual Discovery = 0
3 defaultGatekeeper = 0
4 IPAddress = 0
5 ip = <insert IP address of gatekeeper here>
5 port = 1719
```

If you want to broadcast a general request for a gatekeeper, you can choose to either:

- broadcast to those gatekeepers registered for Multicasting or,
- broadcast a general request to all gatekeepers
DM3 IPLink™ User’s Guide for Windows NT

To broadcast to a Multicast address, there is no need to change the default settings:

```
#Not applicable when manualDiscovery is enabled!
2 rasMulticastAddress = 0
3 ipAddress = 0
4 ip = <224.0.1.41>
4 port = 1718
#Broadcast address
#4 ip = <255.255.255.255>
#4 port = 1718
#station ras port (reply from gatekeeper)
2 rasPort = 1719
```

To broadcast to all gatekeepers, comment lines 4 and 5 and uncomment lines 7 and 8 as shown below:

```
#Not applicable when manualDiscovery is enabled!
2 rasMulticastAddress = 0
3 ipAddress = 0
#4 ip = <224.0.1.41>
#4 port = 1718
#Broadcast address
4 ip = <255.255.255.255>
4 port = 1718
#station ras port (reply from gatekeeper)
2 rasPort = 1719
```

The gateway may identify itself by name and/or by phone number. If you wish to identify your gateway by only one characteristic, comment out the pair of lines that are not used (lines 7/8 or lines 9/10):

```
2 registrationInfo = 0
3 terminalType = 0
4 mc = 0
4 undefinedNode = 0
4 terminal = 0
```
6. Setting IPLink Parameters

3  terminalAlias = 0
4       * = 0
5  h323-ID = "Insert Gateway Name Here"
4       * = 0
5  e164 = 'Insert Gateway Phone Number Here'!

**NOTE:** The config.val syntax requires that the gateway name string be contained within double quotes and the gateway phone number string be contained within single quotes, followed by an exclamation point.
7. Debugging

Several debugging utilities are supplied with the IPLink Development Kit. They are designed to supply information that can help you debug your application.

Debugging can be divided into two basic flows, depending on whether or not the download was successful. Refer to Figure 19. Debug Flow - Download Failed and Figure 22. Debug Flow - Download Succeeded for the basic debug procedure.

Each diagram contains a series of decision and process boxes. Each step of the procedure deals with one possible diagnostic operation that should be performed before continuing to the next step. Several of the steps include extra information contained in “call-outs” as a reminder what should be checked in that particular step.

NOTE: These flow diagrams do not show how to perform the diagnostic procedures in detail. Refer to the description in the following sections for a complete explanation of each diagnostic step.

The following sections describe the diagnostic flow in greater detail using the various utilities supplied with the IPLink platform software.

7.1. Download Failed

If the download failed, use the following debugging procedure:

1. Check the hardware
2. Check the drivers
3. Check the PCD/FCD files

Refer to Figure 19 as a guide to debugging:
Was the download successful?

Yes: See Flow Diagram: Download Succeeded

No: Re-connect all hardware and re-download the DM3 drivers (DCM)

Is the DM3 hardware properly connected?

Yes: Re-connect all hardware properly and re-download the native drivers (DCM)

No: Re-connect all hardware and re-download the DM3 drivers (DCM)

Is the PSTN card set properly?

Yes: 1. Manually start the driver
2. Uninstall DNA and/or MNTI and reinstall

No: Load the default files supplied with the IPLink Development Kit

Are the drivers properly loaded?

Yes: Contact your Dialogic Support Engineer

No: Load the default files supplied with the IPLink Development Kit

Have you changed the PCD or FCD files?

Yes: See Flow Diagram: Download Succeeded

No: Load the default files supplied with the IPLink Development Kit

PSTN board
SCbus cable
Rotary switch = DCM config.
SCBusClock = NOISE

mpd, mcd, Dialogic Configuration Driver,
Dialogic FW Download Drivers, Dialogic SRAM Protocol Driver

Figure 19. Debug Flow - Download Failed
7. Debugging

7.1.1. Check the Hardware

Check that all the hardware is properly connected:

1. Verify that the:
   - DM3 IPLink board is seated properly
   - DM3 boot kernel versions are correct: See the Release Catalog for the correct versions.

Open Dialogic Configuration Manager, double click on the IPT board and select the “Version Info” tab. The boot versions are listed.

![Figure 20. Boot Kernel Versions](image-url)
NOTE: If you cannot verify the boot kernel versions, you may not be able to debug your application. Contact your Dialogic Support Engineer.

2. Verify that the:
   - PSTN board is seated properly
   - Rotary switch on the PSTN board (if so configured) is set to the same value as in the DCM Configuration Manager.
   - PSTN board is set as `SCbusClockMaster = NONE`:
     Open the Dialogic Configuration Manager, double click on the PSTN card you are using and select the “Dialogic Bus” tab. If the `SCbusClockMaster` is set incorrectly, change the Value to “None”.
7. Debugging

3. Verify that the SCbus Cable is connected to both the DM3 IPLink board and the PSTN board.

7.1.2. Check the Drivers

1. Open the Windows Control Panel from the Start Icon.

2. Open the Devices folder. Check that the status of the following drivers is “Running”:
   - MPD
   - MCD
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- Dialogic Configuration Drivers
- Dialogic FW Download Drivers
- Dialogic SRAM Protocol Drivers

Manually start any drivers that are not running. If this fails, uninstall the firmware and then re-install it.

7.1.3. Check the PCD/FCD files

1. Verify that the PCD and FCD files are in your file structure and are pointed to correctly.

2. If you have changed any settings in either of the files, make sure that the changes are correct. Load the default files supplied with the IPLink Development Kit and check for proper operation of the application.
7.2. Download Succeeded

If the download succeeded, use the following debug flow. You will check the output of various utilities. Each step consists of:

- closing the system
- enabling the log file
- re-running the application
- checking the log file

Use the following procedure for debugging your application:

1. Check for H.323 service errors
2. Check for H.323 protocol message errors
3. Check for CP and SP errors

Refer to Figure 22 as a guide to debugging:
Did the application run correctly? 
Yes

1. Close the system 
2. Enable debug info log file 
3. Restart and run the application 

Did you identify a problem with the H.323 service (h323dbg.txt)? 
No

1. Close the system 
2. Enable debug info log file 
3. Restart and run the application 

Did you identify a problem with the H.323 protocol messages (rvtsp1.log)? 
No

1. Close the system 
2. Enable H.323 protocol message log file 
3. Restart and run the application 

Did you find a CP or SP error? 
Yes

1. Close the system 
2. Restart the system 
3. Run DMSTDER 
4. Run the application 

Contact your Dialogic Support Engineer 

No

Contact your Dialogic Support Engineer 

1. Close the system 
2. Enable H.323 protocol message log file 
3. Restart and run the application 

Figure 22. Debug Flow - Download Succeeded

7.2.1. H.323 Service

By default, the log file h323dbg.txt is enabled in the FCD file to print both fatal and non-fatal errors.
7. Debugging

1. Close the system.

2. Re-configure the following parameters to print only fatal errors as described in Chapter 6. Setting IPLink Parameters:
   
   - PrmDebugLevelStack
   - PrmDebugLevelMsg
   - PrmDebugLevelStream
   - PrmDebugLevelStates
   - PrmDebugLevelTimer
   - PrmDebugLevelUtil
   - PrmDebugLevelMNTI

   This reduces the overhead on the CPU and reduces the size of the log file.

   For example, change the PrmDebugLevelStack setting from 2 to 1:

   ! Setting NetTSC_PrmDebugLevelStack parameter to value 2 (Default)
   ! This parameter sets the debug print level for stack module of H323
   ! Values range: 0-4
   ! 0 sets no printouts
   ! 1 sets fatal errors only printouts
   ! 2 adds non-fatal errors printouts
   ! 3 adds warning printouts
   ! 4 adds trace printouts

   Change from 0x2 to 0x1

3. Restart the system and run the application.

4. If the problem doesn’t show up in the log file, return to step 1 and reset the parameters to the next level.

5. Continue until the problem is found. If the problem is not found, continue to the next section.
7.2.2. H.323 Protocol Messages

By default, the log file *rvtsp1.log* is enabled in the FCD file to print fatal and non-fatal errors.

1. Close the system.

2. Re-configure the **PrmDebugLevelRVStack** parameter to print only fatal errors as described in Chapter 6. Setting IPLink Parameters. This reduces the overhead on the CPU and reduces the size of the log file.

   For example, change the PrmDebugLevelRVStack setting from 2 to 1:

   ```
   ! Setting NetTSC_PrmDebugLevelRVStack parameter to value 2 (Default)
   ! This parameter sets the debug print level for RV stack module of H323
   ! Values range: 0-4
   ! 0 sets no printouts
   ! 1 sets fatal errors only printouts
   ! 2 adds non-fatal errors printouts
   ! 3 adds warning printouts
   ! 4 adds trace printouts
   
   }  
   ```

3. Restart the system and run the application.

4. If the problem doesn’t show up in the log file, return to step 1 and reset the parameter to the next level.

5. Continue until the problem is found. If the problem is not found, continue to the next section.

7.2.3. CP and SP Errors

The DM3STDER utility is used to display CP and SP errors.
7. Debugging

NOTE: This utility must be invoked after a successful download, and before running the application.

After successfully downloading the firmware:
1. Open a DOS window.
2. Change to the \PATCH directory.
3. Invoke the DM3STDER utility by typing:
   
   DM3STDER -b <board_number> -f <file_name>
   
   where:
   
   <board_number> is the number of the DM3 board
   <file_name> is a file name for redirecting the data
   
   For example, if there is only a single DM3 board, type:
   
   DM3STDER -b0 -f dm3err.txt
   
4. Run the application.
5. If the problem is not found, contact your Dialogic Support Engineer.

7.3. Information for Customer Support

Have the following information ready when you contact your customer support engineer:

- Log files
- IPT SDK version
- MNTI version
- DNA version
- CP Boot Kernel version
DM3 IPLink™ User’s Guide for Windows NT

- SP Boot Kernel version
- Platform Description
  - CPU type and speed
  - RAM
  - WindowsNT version
- Problem description
The following table should be completed before contacting customer engineering:

<table>
<thead>
<tr>
<th>Description</th>
<th>Version Number/Other Information</th>
</tr>
</thead>
<tbody>
<tr>
<td>IPT SDK</td>
<td></td>
</tr>
<tr>
<td>MNTI</td>
<td></td>
</tr>
<tr>
<td>DNA</td>
<td></td>
</tr>
<tr>
<td>CP Boot Kernel</td>
<td></td>
</tr>
<tr>
<td>SP Boot Kernel</td>
<td></td>
</tr>
<tr>
<td>CPU type and speed</td>
<td></td>
</tr>
<tr>
<td>RAM</td>
<td></td>
</tr>
<tr>
<td>WindowsNT version</td>
<td></td>
</tr>
<tr>
<td>Problem description</td>
<td></td>
</tr>
</tbody>
</table>
Internet Telephony is driving the convergence of the telephone network (PSTN) and the data network (Internet Protocol) into a single communication network that offers powerful, economical, new communications options. Conceptually, Internet telephone gateways provide the following services:

- **On one side, the gateway connects to the telephone world.** It can communicate with any phone in the world. A phone line plugs into the gateway on this end.

- **On the other side, the gateway connects to the Internet world.** It can communicate with any computer in the world that is connected to the Internet.

- **The gateway takes the standard telephone signal,** digitizes it (if it is not already digital), significantly compresses it, packetizes it for the IP, and routes it to a destination over the Internet.

- **The gateway reverses the operation** for packets coming in from the network and going out to the phone.

- **The gateway** also transfers and translates all the call control functions necessary for maintaining the call.

- **Both operations** (coming from and going to the phone network) take place at the same time, allowing a full duplex (two-way) conversation.

The figure below shows a typical configuration, utilizing a single gateway for Phone-to-PC operation and two gateways for Phone-to-Phone operation. Dialogic’s IP Telephony solution provides the ability to configure a system for
both Phone-to-PC and Phone-to-Phone operation, with call control and call information retrieval built-in to the system.

![Diagram of IP Telephony Configuration]

**Figure 23. Typical IP Telephony Configuration**

A number of configurations can be built from this basic operation. Phone-to-PC or PC-to-phone operation can take place with one gateway. Phone-to-phone operation can occur with two gateways.

**Standards**

Support for voice over the Internet, for both PCs and workstations, is rapidly being introduced into the marketplace. The VOIP Service Interoperability Implementation Agreement specifies baseline requirements so that products
Appendix A

complying with its specification can interoperate. The basic standards for Voice over the Internet are as follows:

- All connections are made using the H.323/H.245/H.225/Q.931 session protocol suite.
  Dialogic’s DM3 IPLink platform meets these standards for call establishment and call control. This feature is built-in “under-the-hood” and is invisible to the user. A set of high-level user interface functions simplify the task of creating a host application.

- Connections are made over TCP/UDP/IP protocol layers.
  Dialogic’s IPLink platform makes efficient use of the TCP/UDP/IP protocol layers. A sophisticated packet loss algorithm ensures a high quality of service.

- RTP is used to encapsulate real-time traffic.
  Real-Time Transport Protocol (and its accompanying Real-Time Transport Control Protocol) are built-in to the IPLink platform. Packetization is completely handled by the internal components.

**ITU Recommendation H.323**

Recommendation H.323 describes terminals, equipment, and services for multimedia communication over Local Area Networks (LAN) which do not provide a guaranteed Quality of Service. H.323 terminals and equipment may carry real-time voice, data, and video, or any combination, including videotelephony.

- An **H.323 terminal** is an endpoint on the LAN, capable of providing real-time, two-way communication with another H.323 terminal on another part of the LAN. A terminal, can therefore be either a multi-media PC connected to the LAN, or a gateway such as that described in this document.
The LAN over which H.323 terminals communicate may be a single segment or ring, or it may be multiple segments with complex topologies. The operation of H.323 terminals over the multiple LAN segments (including the Internet) may result in poor performance. The possible means by which quality of service might be assured on such types of LANs/internetworks is beyond the scope of this recommendation.

H.323 terminals may be integrated into personal computers or implemented in stand-alone devices such as videotelephones. Support for voice is mandatory, while data and video are optional, but if supported, the ability to use a specified common mode of operation is required, so that all terminals supporting that media type can interwork. H.323 allows more than one channel of each type to be in use. Other Recommendations in the H.323 series include H.225.0 packet and synchronization, H.245 control, H.261 and H.263 video codecs, G.711, G.722, G.728, G.729, and G.723.1 audio codecs, and the T.120 series of multimedia communications protocols.

H.323 makes use of the logical channel signaling procedures of Recommendation H.245, in which the content of each logical channel is described when the channel is opened. Procedures are provided for expression of receiver and transmitter capabilities, so transmissions are limited to what receivers can decode, and so that receivers may request a particular desired mode from transmitters.

The following elements are mandatory within the scope of H.323:

- The **Audio Codec** encodes the audio signal from the microphone for transmission and decodes the received audio code which is output to the loudspeaker.

H.323 specifies a number of audio codecs (G.711, G.722, G.723.1, MPEG1, G.728, and G.729). The standard allows the parties to negotiate the codec. Several of the specified codecs are especially relevant for Internet use. Other
codecs can be added, allowing proprietary codecs to be integrated into an Internet Gateway. Dialogic’s IPLink platform supports multiple codecs, both standard and proprietary.

- **System Control Unit** (H.245, H.225.0) provides signaling for proper operation of the H.323 terminal. It provides for call control, capability exchange, signaling of commands and indications, and messages to open and fully describe the content of logical channels.

  This capability is built-in to the Dialogic IPLink platform. The application developer can easily access these features via high-level functions.

- **H.225.0 Layer** (H.225.0) formats the transmitted audio data and control streams into messages for output to the network interface, and retrieves the received audio and control streams from messages which have been input from the network interface. In addition, it performs logical framing, sequence numbering, error detection, and error correction.

  H.225.0 makes use of **RTP/RTCP** (Real-time Transport Protocol/Real-Time Transport Control Protocol) for media stream packetization and synchronization for all underlying LANs.

  This capability is built-in to the Dialogic IPLink platform. The application developer can easily access these features via high-level functions.

- **LAN Interface** is implementation specific and is outside the scope of Recommendation H.323. However, the LAN interface shall provide the services described in Recommendation H.225.0. This includes:
  - Reliable (e.g., TCP, SPX) end-to-end service is mandatory for the H.245 Control Channel and the Call Signaling Channel.
  - Unreliable (e.g., UDP, IPX) end-to-end service is mandatory for the Audio Channels.
These services may be duplex or simplex, unicast or multicast depending on the application, the capabilities of the terminals, and the configuration of the LAN.

Dialogic’s IPLink platform supports the reliable channels and ensures a high quality of service on the unreliable channel (UDP).
Appendix B

config.val File

# H.323 Stack Configuration File

1 system = 0
2 pdlName = 'radvh323.raw'
3 allocations = 0
3 vtPoolSize = 100000
3 vtNodeCount = 20000
3 channels = 500
3 chanDescs = 5
3 messages = 200
3 nameChans = 10
3 tpktChans = 80
3 udpChans = 5
3 protocols = 400
3 maxProcs = 400
3 maxDepth = 200

1 RAS = 0
2 responseTimeOut = 20
2 gatekeeper = 1
2 manualRAS = 1
2 manualDiscovery = 0
3 defaultGatekeeper = 0
4 ipAddress = 0
4 ip = <200.200.200.200>
5 port = 1719

2 registrationInfo = 0
3 terminalType = 0
4 mc = 0
4 undefinedNode = 0
4 terminal = 0
3 terminalAlias = 0
4 * = 0
5 h323-ID = 'Name'
4 * = 0
5 e164 = '1111'

#Not applicable when manualDiscovery is enabled!
2 rasMulticastAddress = 0
3 ipAddress = 0
4 ip = <224.0.1.41>
4 port = 1718
#Broadcast address
#4 ip = <255.255.255.255>
#4 port = 1718
#station ras port (reply from gatekeeper)
2 rasPort = 1719

1 Q931 = 0
2 responseTimeOut = 50
2 connectTimeOut = 500
2 callSignalingPort = 1720
2 maxCalls = 10
#2 notEstablishControl = 0
2 manualAccept = 1
1 h245 = 0
2 masterSlave = 0 # Sequence <Optional>
3 terminalType = 70 # INTEGER [0..255]
#3 manualOperation = 0 # NULL
3 timeout = 40

2 capabilities = 0 # Sequence <Optional>
3 timeout = 40
#3 manualOperation = 0
3 terminalCapabilitySet = 0 # Sequence
4 sequenceNumber = 0 # INTEGER [0..255]
4 protocolIdentifier = [00] # Object Identifier [0..0]

4 multiplexCapability = 0
5 h2250Capability = 0 # Sequence
6 maximumAudioDelayJitter = 60 # INTEGER [0..1023]
6 receiveMultipointCapability = 0 # Sequence
7 multicastCapability = 0 # BOOLEAN [0..0]
7 multiUniCastConference = 0 # BOOLEAN [0..0]
7 mediaDistributionCapability = 0
8 * = 0 # Sequence
9 centralizedControl = 0 # BOOLEAN [0..0]
9 distributedControl = 0 # BOOLEAN [0..0]
9 centralizedAudio = 0 # BOOLEAN [0..0]
9 distributedAudio = 0 # BOOLEAN [0..0]
9 centralizedVideo = 0 # BOOLEAN [0..0]
9 distributedVideo = 0 # BOOLEAN [0..0]
6 transmitMultipointCapability = 0 # Sequence
multicastCapability = 0 # BOOLEAN [0..0]
multiUnicastConference = 0 # BOOLEAN [0..0]
mediaDistributionCapability = 0

* = 0 # Sequence

centralizedControl = 0 # BOOLEAN [0..0]
distributedControl = 0 # BOOLEAN [0..0]
centralizedAudio = 0 # BOOLEAN [0..0]
distributedAudio = 0 # BOOLEAN [0..0]
centralizedVideo = 0 # BOOLEAN [0..0]
distributedVideo = 0 # BOOLEAN [0..0]

receiveAndTransmitMultipointCapability = 0 # Sequence
multicastCapability = 0 # BOOLEAN [0..0]
multiUnicastConference = 0 # BOOLEAN [0..0]
mediaDistributionCapability = 0

* = 0 # Sequence

centralizedControl = 0 # BOOLEAN [0..0]
distributedControl = 0 # BOOLEAN [0..0]
centralizedAudio = 0 # BOOLEAN [0..0]
distributedAudio = 0 # BOOLEAN [0..0]
centralizedVideo = 0 # BOOLEAN [0..0]
distributedVideo = 0 # BOOLEAN [0..0]

mcCapability = 0 # Sequence

centralizedConferenceMC = 0 # BOOLEAN [0..0]
decomcentralizedConferenceMC = 0 # BOOLEAN [0..0]
rtcpVideoControlCapability = 0 # BOOLEAN [0..0]
mediaPacketizationCapability = 0 # Sequence
h261aVideoPacketization = 0 # BOOLEAN [0..0]

capabilityTable = 0

* = 0 # Sequence
capabilityTableEntryNumber = 1 # INTEGER [1..65535]
capability = 0
receiveAudioCapability = 0
g711Ulaw64k = 30 # INTEGER [1..256]

* = 0 # Sequence
capabilityTableEntryNumber = 2 # INTEGER [1..65535]
capability = 0
receiveAudioCapability = 0
g7231 = 0
maxAl-sduAudioFrames = 1 # INTEGER [1..256]
silenceSuppression = 1 # INTEGER [1..256]

capabilityDescriptors = 0

* = 0 # Sequence
capabilityDescriptorNumber = 0 # INTEGER [0..255]
simultaneousCapabilities = 0
* = 0
* = 1 # INTEGER [1..65535]
* = 2 # INTEGER [1..65535]

channels = 0

* = 0 # Sequence
name = 'g711Ulaw10' # IA5String [1..128]
dataType = 0
audioData = 0
g711Ulaw64k = 10 # INTEGER [1..256]

* = 0 # Sequence
name = 'g711Ulaw20' # IA5String [1..128]
dataType = 0
audioData = 0
g711Ulaw64k = 20 # INTEGER [1..256]

* = 0 # Sequence
name = 'g711Ulaw30' # IA5String [1..128]
dataType = 0
audioData = 0
g711Ulaw64k = 30 # INTEGER [1..256]

* = 0 # Sequence
name = 'g7231' # IA5String [1..128]
dataType = 0
audioData = 0
g7231 = 0
maxAl-sduAudioFrames = 1 # INTEGER [1..256]
silenceSuppression = 1 # INTEGER [1..256]
Appendix C

ipt.fcd File

;;; FCD File generated from: qvs_t1.config
;;; And appended to for the IPT NetTSC
;;; Search down for [NetTSC] to see beginning of additions
;;; which were typed in (not generated)
;;; Note that for each ["component"]
;;; [SendMsg] sends the parameter
;;; and then [RcvMsg] verifies that it was received

************************************************************
* [NOTE: This file illustrates only the IP Telephony parameters.]  *
************************************************************
*
[NetTSC]
{
  Attribute: 0:0x1e
  ! NetTSC_PrmDebugLevelStack
  (0=OFF, 1=FATAL, 2=ERROR, 3=Warning, 4=Info)
  ; SetParam (7694, 2) on NetTSC
  [SendMsg]
  {
    ! Setting parameter for NetTSC.0
    MsgType: 0x8
    UInt32: 0x1e0e
    UInt32: 0x2
  }
  [RcvMsg]
  {
    ! Expecting parm Complete from NetTSC.0
    MsgType: 0x9
    TimeOut: 0x1388
  }
}
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  UInt32  : 0x1e0f ;7695
  UInt32  : 0x2 ;2
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
}

! NetTSC_PrmDebugLevelStream
(0=OFF,1=FATAL,2=ERROR,3=Warning,4=Info)
;; SetParam (7696,0) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  UInt32  : 0x1e10 ;7696
  UInt32  : 0x0 ;0
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
}

! NetTSC_PrmDebugLevelStates
(0=OFF,1=FATAL,2=ERROR,3=Warning,4=Info)
;; SetParam (7697,4) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  UInt32  : 0x1e11 ;7697
  UInt32  : 0x4 ;4
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
Appendix C

TimeOut : 0x1388 ;5000
}

! NetTSC_PrmDebugLevelTimer
(0=OFF,1=FATAL,2=ERROR,3=Warning,4=Info)
;; SetParam (7698,0) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
   MsgType : 0x8 ;8
   UInt32 : 0x1e12 ;7698
   UInt32 : 0x0 ;0
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
   MsgType : 0x9 ;9
   TimeOut : 0x1388 ;5000
}

! NetTSC_PrmDebugLevelUtil
(0=OFF,1=FATAL,2=ERROR,3=Warning,4=Info)
;; SetParam (7699,2) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
   MsgType : 0x8 ;8
   UInt32 : 0x1e13 ;7699
   UInt32 : 0x2 ;2
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
   MsgType : 0x9 ;9
   TimeOut : 0x1388 ;5000
}

! NetTSC_PrmDebugLevelMNTI (0-3=OFF,4=Info)
;; SetParam (7700,0) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
   MsgType : 0x8 ;8


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```
(UInt32 : 0x1e14 ; 7700
UInt32 : 0x0 ; 0
)
[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ; 9
TimeOut : 0x1388 ; 5000
}

! NetTSC_PrmDebugLevelRVSTACK (0-3=OFF,4=Info)
;;; SetParam (7710,0) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
MsgType : 0x8 ; 8
UInt32 : 0x1e1e ; 7710
UInt32 : 0x0 ; 0
}
[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ; 9
TimeOut : 0x1388 ; 5000
}

! NetTSC_PrmDialogicEnable (0=Standard Gateway,1=Dialogic Gateway)
;;; SetParam (7705,1) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
MsgType : 0x8 ; 8
UInt32 : 0x1e19 ; 7705
UInt32 : 0x1 ; 1
}
[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ; 9
TimeOut : 0x1388 ; 5000
}
```
! prmECActive (0=Enable, 1=Disable)
;; SetParam (6930, 0) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  Uint32 : 0x1b12 ;6930
  Uint32 : 0x1 ;0
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
}

! prmECOrder (128/256)
;; SetParam (6931, 128) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  Uint32 : 0x1b13 ;6931
  Uint32 : 0x80 ;128
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
}

! prmVolumeControl (1-8)
;; SetParam (6932, 2) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  Uint32 : 0x7fffffff ;.999999
  Uint32 : 0x0 ;0
}
[RcvMsg]
{}
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
{}
! prmSQM_1
;; SetParam (6919,2) on NetTSC
[RcvMsg]
{}
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  UInt32  : 0x1b07 ;6919
  UInt32  : 0x6 ;6
{}
[RcvMsg]
{}
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
{}
! prmSQM_2
;; SetParam (6918,2) on NetTSC
[RcvMsg]
{}
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
  UInt32  : 0x1b08 ;6918
  UInt32  : 0xf ;15
{}
[RcvMsg]
{}
! Expecting parm Complete from NetTSC.0
  MsgType : 0x9 ;9
  TimeOut : 0x1388 ;5000
{}
! prmSQM_3
;; SetParam (6927,2) on NetTSC
[RcvMsg]
{}
! Setting parameter for NetTSC.0
  MsgType : 0x8 ;8
The image contains a document page with programming code. Here is the plain text representation of the code:

```plaintext
UInt32 : 0x1b0f ;6927
UInt32 : 0x78 ;120

}[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ;9
TimeOut : 0x1388 ;5000
}

! prmSQM_4
;; SetParam (6928,100) on NetTSC
[RcvMsg]
{
! Setting parameter for NetTSC.0
MsgType : 0x8 ;8
UInt32 : 0x1b10 ;6928
UInt32 : 0x3c ;60
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ;9
TimeOut : 0x1388 ;5000
}

! prmSQM_5
;; SetParam (6918,0) on NetTSC
[RcvMsg]
{
! Setting parameter for NetTSC.0
MsgType : 0x8 ;8
UInt32 : 0x1b06 ;6918
UInt32 : 0x2 ;2
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
MsgType : 0x9 ;9
TimeOut : 0x1388 ;5000
}

! prmSQM_6
;; SetParam (6917,2) on NetTSC
```
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType       : 0x8    ;8
  UInt32        : 0x1b05    ;6917
  UInt32        : 0x4    ;4
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType       : 0x9    ;9
  TimeOut       : 0x1388    ;5000
}

! prmSQM_7
;;; SetParam (6933,1) on NetTSC
[SendMsg]
{
! Setting parameter for NetTSC.0
  MsgType       : 0x8    ;8
  UInt32        : 0x1b15    ;6933
  UInt32        : 0x1    ;1
}

[RcvMsg]
{
! Expecting parm Complete from NetTSC.0
  MsgType       : 0x9    ;9
  TimeOut       : 0x1388    ;5000
}
)
Glossary

Asynchronous mode An operating mode of certain DM3 kernel or host library function calls. Asynchronous mode is a non-blocking mode that is typically used when a function involves operation on a remote processor (e.g. a host library function that initiates a kernel operation on a DM3 platform’s Control Processor) or when data movement via the MMA and global memory is required. In asynchronous mode a value signifying “pending” status is immediately returned to the calling function, and the actual completion or failure of the operation is reported to the caller via a DM3 result message.

Cluster A collection of DM3 components that share a set of TDM timeslots on the network interface or the SCbus. Components are bound to a particular cluster and its assigned timeslots in an allocation operation. The component that contains the ports to the TDM timeslots (generally the first component allocated into the cluster) is called the central component.

Codec see COder/DECoder

COder/DECoder A circuit used on Dialogic boards to convert analog voice data to digital and digital voice data to analog audio

Component An executable entity that resides on a particular processor on the DM3 platform and provides a single type of service to a DM3 resource. For example, a decoder component may be part of the Player resource, residing on a signal processor (SP) and providing specific decoding services for mediastream playback. Currently, the object code for a DM3 component is linked with the DM3 and RTOS kernels and downloaded to the target processor as a single Processor Load Module (PLM). The DM3 architecture supports multiple instances of components to allow independent simultaneous handling of multiple media streams.

Component instance A logical entity that represents a single thread of control for the operations associated with a DM3 component. Each DM3 component supports multiple instances up to some defined
maximum number (which may be greater than the number of instances that may actually be implemented with existing platform hardware). Instances are addressable units, and DM3 messages may be sent to individual instances of a component. An instance generally corresponds to an individual media stream, and when an instance is active it is usually bound to a particular set of SCbus data timeslots and to a SCSA Group.

**Computer Telephony (CT)**  Adding computer intelligence to the making, receiving, and managing of telephone calls.

**Control Processor (CP)**  A processor that is present on the motherboard of every DM3 platform for the management of the SCbus, for host communication configuration, and to implement some of the features of DM3 Resources. All current DM3 platforms use an Intel i960 as the Control Processor.

CP  See Control Processor

CT  See Computer Telephony

**DM3 address**  A bitmapped data structure that identifies a specific DM3 software entity and the processor, board, and/or node where the entity resides. DM3 addresses are used to identify components, component instances, clusters, and tasks, and partially specified addresses may be used to identify the board or processor. Each DM3 message contains source and destination component addresses, and many function calls require one or more component addresses as arguments.

**DM3 architecture**  A platform architecture for a family of high density, high performance, call processing products. The architecture specifies both the hardware platform and the firmware modules in such a way that resources are highly portable across multiple platform implementations, including VME, PCI, and CompactPCI variants.

**DM3 kernel**  Firmware that is present on each processor on a DM3 platform to support configuration management, host communication, inter processor communication and control, and SCSA firmware services among others. There may be more than one version of the DM3 kernel firmware depending on the type of processor (e.g. CP vs. DSP or RISC signal processor) and the intended application of each processor.
**DM3 load module**  A loadable block of executable firmware for a particular processor on a DM3 platform. A DM3 Resource may be packaged as one or more DM3 load modules.

**DM3 message**  A formatted block of data that is passed between the host and a DM3 component or between two DM3 components via the global memory and MMA ASIC. All DM3 messages have a packed, fixed-format header that contains transport information (source and destination addresses, priority, amount of data in the message body, etc.) and an arbitrary transaction ID number as well as the message type, which is the primary content of the message. The DM3 software environment provides macros to put values into and get values from the packed message header independent of the processor word length and endianness. In addition to the fixed-format header, messages may optionally have a body that contains additional data in typed fields that are accessed using macros or low-level function calls.

**DM3 platform**  A specific DM3 hardware implementation (a board) for a particular system bus, which includes a control processor (CP), optional signal processors (SPs), and optional peripherals (e.g. communication daughterboard for development or DNI daughterboard for additional network interfaces).

**DM3 signal computing resource**  Also “DM3 technology resource” or simply “DM3 resource”. A software component which may be deployed on compatible platforms based upon the DM3 architecture. A resource is a logical entity that supports a number of instances of a fixed set of features. The DM3 resource maps closely, though not exactly, with the SCSA Resource paradigm. For example, a DM3 Player resource may be a DM3 resource providing playback capability; it may carry out some or all of the functionality defined in the SCSA Player API. A DM3 Resource is typically made up of a DM3 host-side library and one or more DM3 firmware resource components (or simply “DM3 components”).

**DM3 task**  An executable entity which is managed and scheduled by the DM3 kernel and underlying real-time operating system that is present on all of the processors on the DM3 platform. Each task maintains its own environment and has its own stack. Each DM3 task implements a message queue that services one or more components or component instances.
DTMF  See Dual-Tone Multi-Frequency

Dual-Tone Multi-Frequency  A way of signaling consisting of a push-button or touch-tone dial that sends out a sound consisting of two discrete tones that are picked up and interpreted by telephone switches (either PBXs or central offices).

Endpoint  An H.323 Gateway, CMA client, LDAP server, or CMA server. An endpoint can call and be called. It generates and/or terminates information streams.

FCD file  An ASCII file that lists any non-default parameter settings that are necessary to configure a DM3 hardware/firmware product for a particular feature set. The downloader utility reads this file, and for each parameter listed generates and sends the DM3 message necessary to set that parameter value.

Feature Configuration File  See FCD file

Frame  1) A set of SCbus timeslots which are grouped together for synchronization purposes. The period of a frame is fixed (at 125 µsec) so that the number of time slots per frame depends on the SCbus data rate.

2) In the context of DSP programming (e.g. DM3 component development), the period defined by the sample rate of the signal data.

Gatekeeper  An H.323 entity on the Internet that provides address translation and control access to the network for H.323 Terminals and Gateways. The Gatekeeper may also provide other services to the H.323 terminals and Gateways, such as bandwidth management and locating Gateways.

Gateway  An H.323 Gateway (GW) is an endpoint on the Internet which provides for real-time, two-way communications between H.323 Terminals on the Network and other ITU Terminals on a wide area network, or to IP Telephony clients.

H.323  The specification that defines packet standards for terminals, equipment, and services for multimedia communications over LANs. Adopted by the Internet telephony community as the standard for communicating over any packet network, including the Internet.
**International Telecommunications Union (ITU)**  
An organization established by the United Nations to set telecommunications standards, allocate frequencies to various uses, and hold trade shows every four years.

**Internet**  
An inter-network of networks interconnected by bridges or routers. LANs described in H.323 may be considered part of such inter-networks.

**Internet Protocol (IP)**  
The TCP/IP standard protocol that defines the IP datagram as the unit of information passed across an internet and provides the basis for connectionless, best-effort packet delivery service.

**Internet Service Provider (ISP)**  
A vendor who provides direct access to the Internet.

**Internet Telephony**  
Technology that lets you make voice phone calls over the Internet or other packet networks using your PC, via gateways and standard telephones.

**IP**  
See Internet Protocol

**ISP**  
See Internet Service Provider

**ITU**  
See International Telecommunications Union.

**Parameter**  
A type of datum that is passed to or from a component via a DM3 message. Parameters generally have two elements, a type and a specific value. Parameters passed to a component affects the operational characteristics of the component, and parameters passed from a component can be used to report the operating state of the component. As an example, the standard Player component accepts a parameter called ParmDuration, which specifies the length of the file that is being played back, and can report its current state (e.g. playing, paused, etc.) via a parameter called ParmState.

**PCD file**  
An ASCII text file that contains product or platform configuration description information that is used by the DM3 downloader utility program. Each of these files identifies the hardware configuration and firmware modules that make up a specific hardware/firmware product. Each type of DM3-based product used in a system requires a product-specific PCD file.
**Reliable Channel** A transport connection used for reliable transmission of an information stream from its source to one or more destinations.

**Reliable Transmission** Transmission of messages from a sender to a receiver using connection-mode data transmission. The transmission service guarantees sequenced, error-free, flow-controlled transmission of messages to the receiver for the duration of the transport connection.

**Resource** A conceptual entity defining a feature set, which can be implemented by writing firmware components that run on one or more processors on a DM3 platform. A simple example of a resource is a player, which is composed of a player component, which runs on the Control Processor of a DM3 platform and handles control and communications functions, plus a decoder component which typically runs on a Signal Processor and processes the encoded data stream that is being played back.

**RTCP** Real Time Control Protocol

**RTP** Real Time Protocol

**SCbus** The standard bus for communication within a SCSA node. The architecture of SCbus includes a 16-wire TDM data bus that operates at 2, 4 or 8 Mbps and a serial message bus for control and signaling. DM3 platforms provide an SCbus interface for interconnection of multiple DM3 platforms, or connection to other SCSA-compatible hardware. The DM3 platform supports timeslot bundling for high bandwidth, and can access up to 256 of the 2048 SCbus timeslots via two SC4000 ASICs.

**Signal Processor (SP)** An embedded processor on a DM3 platform that is used to execute signal processing algorithms. The DM3 architecture supports multiple types of SPs, including RISC processors (e.g. PowerPC) as well as general purpose DSPs (e.g. Motorola 5630x), and platforms may be configured with a mixed complement of SP types.

**Standard message set** A predefined set of DM3 messages which handle functionality that is common to all DM3 component, such as setting and getting parameters and managing RTC events. Each DM3 component should support all of the standard messages that are relevant to the component’s operation. Each component will generally
supplement the standard message set with its own set of proprietary or component-specific messages.

**Synchronous mode**  An operating mode of certain DM3 kernel or host library function calls. Synchronous mode is a blocking mode that is typically used only when a function executes on the local processor using data that is also local to the processor (i.e. that does not require any data movement via the MMA).

**TCP**  see Transmission Control Protocol

**Terminal**  An H.323 Terminal is an endpoint on the local area network which provides for real-time, two-way communications with another H.323 terminal, Gateway, or Multipoint Control Unit. This communication consists of control, indications, audio, moving color video pictures, and/or data between the two terminals. A terminal may provide speech only, speech and data, speech and video, or speech, data, and video.

**Transmission Control Protocol**  The TCP/IP standard transport level protocol that provides the reliable, full duplex, stream service on which many application protocols depend. TCP allows a process on one machine to send a stream of data to a process on another. It is connection-oriented in the sense that before transmitting data, participants must establish a connection.

**UDP**  see User Datagram Protocol

**User Datagram Protocol**  The TCP/IP standard protocol that allows an application program on one machine to send a datagram to an application program on another machine. Conceptually, the important difference between UDP datagrams and IP datagrams is that UDP includes a protocol port number, allowing the sender to distinguish among multiple destinations on the remote machine.
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