Description

DCS-816

DIGITAL COMMUNICATION SYSTEM

for Europe

GENERAL DESCRIPTION

SAMSUNG

SAMSUNG ELECTRONICS CO. LTD.

Publication Information

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PART 1. SYSTEM OVERVIEW

1.1 INTRODUCTION

DCS-816, Digital Communication System, is a digital telephone system designed for small to medium-sized businesses. The system employs the very latest Digital Signal Processors (DSP) technology.

DCS-816 is easy to install and maintain, convenient to use and provides a variety of features, including advanced features such as ISDN and CTI.

DCS-816 provides some interface cards that allow connection to the public telephone network or private networks. These are generally referred to as trunk cards. Besides the trunk cards, it provides some option cards for serial interfaces and the Auto Attendant feature. Two types of telephones can be connected to the system: proprietary digital phones called "keysets" connect to digital line interface (DLI) ports; standard telephone sets, generally called "single line telephones," connect to single line interface (SLI) ports. In addition, DLI station ports are used to connect peripheral devices such as door phone interface modules (DPIMs) and CTI interface modules.

Miscellaneous circuits are provided to allow such optional features as external paging, music on hold, background music, general-purpose dry contact and emergency power failure transfer. All DCS keysets utilise a single PCB with surface-mounted components assuring the highest product quality and long life. Samsung's customary large, easy-to-read displays and LEDs in the button (key) design make them much easier to use. In many instances, sophisticated features are made simple through the use of friendly display prompts or push-on/push-off feature keys.

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NOTE:

1. It is not possible to expand from DCS-816 to other DCS systems (e.g. DCS, DCS Compact).

2. DCS-816 cannot accommodate cards from other Samsung digital systems **except** for the DCS Compact BRI card.

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1.2 SIZE AND CONFIGURATION

The **DCS-816** system consists of the Basic Key Service Unit (KSU), option cards, trunk interface cards, interface modules and digital keysets. The KSU is a single metal cabinet, and in its basic form has 16 station ports: twelve (12) keyset ports and four (4) single line telephone (SLT) ports.

Additionally, the KSU contains one (1) trunk expansion slot, two (2) option connections and miscellaneous circuits: one (1) external page Interface, one (1) general-purpose dry contact and one (1) music interface selectable to internal built-in music (a melody chip) or external music.

Two types of telephones can be connected to the system: proprietary digital stations called "keysets" that connect to digital line interface (DLI) ports and standard telephone sets, generally called "single line telephones", that connect to single line interface (SLI) ports.

The Trunk Expansion slot provides an option to select an analogue trunk card or a BRI card.

The DCS-816 system also has some option cards to provide serial I/O ports or AA facilities. The Basic KSU has two (2) analogue trunk ports and 16 stations, and the trunk expansion slot provides a maximum capacity of 10 trunk ports and 16 stations.

The system maximum capacity is as follows:

Stations	**16 (12 keysets, 4 SLTs)**
ISDN channels	**8 (4 BRI ports)**
Analogue C.O. lines	**10**
Music channel	**1 (Internal or External)**
External Page	**1**
General-purpose	**Dry Contact 1**
Serial I/O ports	**2 (DB9 connectors)**
Auto Attendant channels	**4**

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CONFIGURATION NOTES
1. No KDbs (keyset daughter-boards for additional DLI or SLI expansion) can be installed.
2. Only one (1) trunk expansion card can be installed.
3. Only one SIO option card and one AA option card can be installed.
4. The system cannot accommodate cards from other Samsung digital systems **except** for the DCS Compact BRI card.
5. Every analogue trunk interface contains an interface for MPD or PRS which is optional. The MPD or PRS hybrid chips of the DCS system can be used.

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The DCS-816 system has only one expansion slot dedicated to a trunk expansion card.

**Trunk Expansion
Card Installed**

System Configuration

4TRK

616 System (Analogue Trunk)

4 Analogue C.O. Lines (Expansion Card)
2 Analogue C.O. Lines (Basic KSU)
12 Keysets (Basic KSU)
4 Single Line Telephones (Basic KSU)

8TRK

1016 System (Analogue Trunk)

8 Analogue C.O. Lines (Expansion Card)
2 Analogue C.O. Lines (Basic KSU)
12 Keysets (Basic KSU)
4 Single Line Telephones (Basic KSU)

2BRI

616 System (Analogue/Digital Trunk)

2 BRI ports (4 channels)
2 Analogue C.O. Lines (Basic KSU)
12 Keysets (Basic KSU)
4 Single Line Telephones (Basic KSU)

4BRI

1016 System (Analogue/Digital Trunk)

4 BRI ports (8 channels)
2 Analogue C.O. Lines (Basic KSU)
12 Keysets (Basic KSU)
4 Single Line Telephones (Basic KSU)

DCS-816 System Configuration Table   **Figure 1-1**
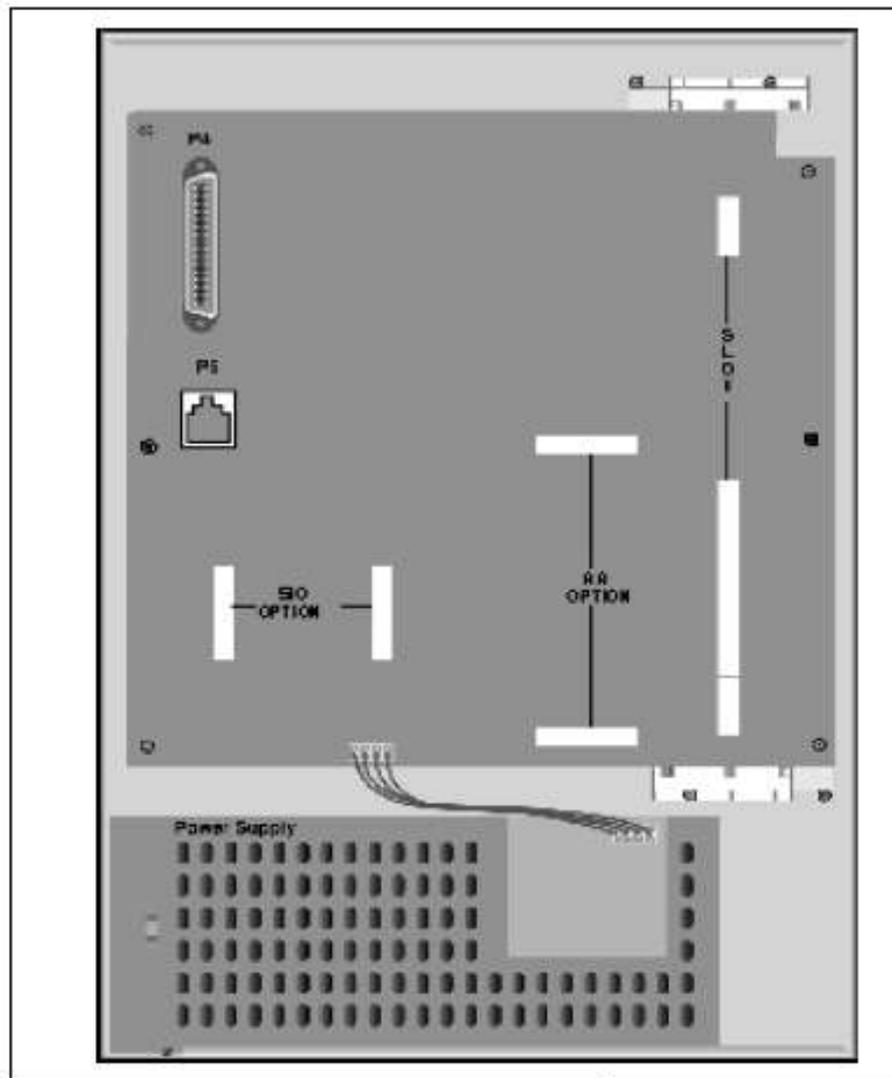
Figure 1-2
DCS-816 System Configuration

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1.3 TECHNOLOGY

SWITCHING

System switching is accomplished by means of a custom IC "engine" that provides 256 switchable digital channels. The engine is controlled by its own 16-bit Motorola™ MC 68000 microprocessor and switching control program. Each of the 256 digital channels is automatically assigned to carry voice or data as required by system operation in a PCM format.

In addition to the 256 channels mentioned above, the system also utilises Digital Signal Processors (DSPs). Each DSP may be configured by the switching control program as a DTMF sender, a DTMF receiver or as a C.O. tone detector on a per-call basis. DCS-816 is equipped with one engine chip containing four (4) DSP channels.

The DSP channels are fully shared throughout the system as a common resource.

MEMORY

The systems operate using stored program control. This program is stored in EPROM chips. All specific customer data is stored in random access memory (RAM) which is protected against the loss of AC power to the system by an on-board battery backup system.

MICROPROCESSORS

DCS-816 uses distributed processing. The primary system processor is a 16-bit Motorola™ MC68000 operating at a clock speed of 8 MHz. Secondary processing is done in some special cards, such as BRI, and also in the keysets. The digital keyset uses a Hitachi H8 processor for data communication within the DCS-816.

1.4 PROGRAMMING

DCS-816 is self-configuring. This means that when the power is switched on, the system reads the types and locations of all installed cards and telephones and assigns default data to them. This data provides for system operation within seconds after power is switched on. All trunks and stations are assigned according to the default numbering plan. This numbering plan is flexible and may be changed to suit customer requirements. The installing technician customises these default data to meet the end user's requirements.

The systems can be programmed from any LCD display keyset without interrupting system operation. There are three levels of programming: technician (or system), customer and station.

The technician (system) level has access to all programs and can allow the customer access to system programs as needed. Technician and customer access is controlled by different security passcodes and access procedures.

DCS-816 also allows the use of a proprietary computer program called PCMMC. This permits a technician to program the system using a personal computer. PCMMC can be used on-site to modify the customer database or to download (save) the entire customer database to a file. This file can then be saved as a backup and be uploaded when required to restore the database.

Through the use of modems, PCMMC can access a DCS-816 system remotely (off-site) to make database changes or perform uploads or downloads of the customer database as if the technician were on-site.

PART 2. DCS-816 - HARDWARE DESCRIPTIONS

2.1 Basic Key Service Unit

The DCS-816 Key Service Unit (KSU, Figure 2-1) is a single metal cabinet containing the following:

- A power supply

- Processing, memory and switching: CPU (68EC000), 2 Mbyte program memory,

- 256 Kbyte data memory with a backup super capacitor, real-time clock

- Two (2) analogue C.O. line interfaces with connections for PRS or MPD chips.

- Twelve (12) digital keyset interfaces

- Four (4) single line telephone interfaces with loop-disconnect facilities

- One (1) Internal Music source or an External Music interface for Music-on-Hold and Background Music

Figure 2-1

- One (1) External Page interface

- One (1) general-purpose dry contact

- Two (2) Power Failure Transfer relays. Two (2) pre-assigned analogue trunk lines will be automatically transferred to station 15 and station 16 respectively in case of power failure. Which analogue trunk lines are to be transferred must be decided by the installer

- A 1.5W sinusoidal ring generator for single line telephones

- One (1) trunk expansion slot, one (1) SIO option connection for SIO1 and one (1) AA option connection

- One (1) champ connector and one (1) RJ-45 connector for external connections and other connectors for the Power Supply Unit and the external backup battery.
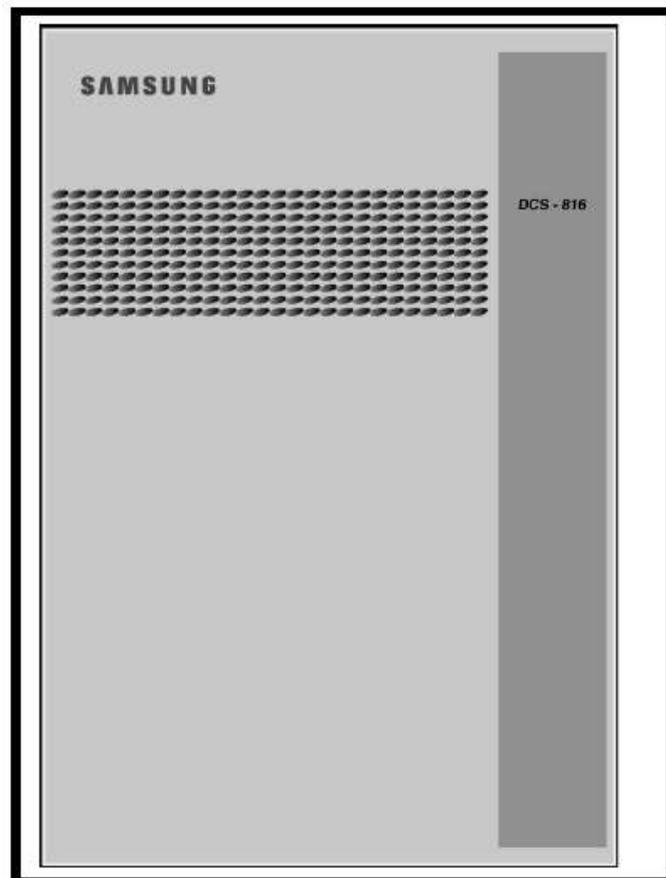
2.2 OPTION CARDS

SIO Option Card
Two dedicated connectors in the Basic KSU are for the SIO option card. The SIO1 option card contains two (2) RS232C serial interfaces with DB9 connectors for connection to a personal computer, a printer or a modem. They are for SMDR or PCMMC. One per system.

AA Option Card
The AA option card is equipped with 4-channel built-in AA. There are two dedicated connectors in the Basic KSU for the option card. The card has its own processor, 68EC000, 64 Kbyte program memory, 512 Kbyte pre-recorded message memory, 1 Mbyte data and customer message memory and a DSP chip for AA processing. One card per system.

2.3 TRUNK EXPANSION CARDS

Only one slot is available for trunk expansion.

4TRK Card
The 4TRK card provides four (4) analogue C.O. interfaces. Each interface contains an interface for MPD or PRS, which is optional. The MPD or PRS hybrid chips from the DCS system can be used.

8TRK Card
The 8TRK card provides eight (8) analogue C.O. interfaces. Each interface contains an interface for MPD or PRS, which is optional. The MPD or PRS hybrid chips from the DCS system can be used.

2BRI Card
The DCS-816 system accommodates the 2BRI card from the DCS Compact system, which contains two (2) basic rate access ports, i.e. four (4) ISDN channels. Protocol is compatible with Euro-ISDN BRI.

4BRI Card
The DCS-816 system accommodates the 4BRI card from the DCS Compact system, which contains four (4) basic rate access ports, i.e. eight (8) ISDN channels. Protocol is compatible with Euro-ISDN BRI.

NOTE: DCS-816 cannot accommodate interface cards from other DCS systems (i.e. DCS, DCS Compact I, or DCS Compact II) **except** for the DCS Compact BRI cards.

2.4 SPECIFICATIONS

The following tables provide technical data for the DCS-816 hybrid/key telephone system.

ELECTRICAL SPECIFICATIONS

AC INPUT	220 VAC $\pm$ 20%, 48-63 Hz
POWER CONSUMPTION (MAX)	60 WATTS MAX
DC OUTPUT	FUSE RATING 2.0 AMP
	+5 VOLTS 1.5 AMPS MAX
	-5 VOLTS 0.5 AMPS MAX (on the base board)
	-55 VOLTS 0.7 AMPS MAX
	-54 VOLTS 0.4 AMPS MAX

DIMENSIONS AND WEIGHTS

	HEIGHT (mm)	WIDTH (mm)	DEPTH (mm)	WEIGHT (kg)
BASIC KSU	465	315	120	
DIGITAL KEYSET (12/24B MODELS)	108	216	229	1.2
DIGITAL KEYSET (DS-2100B)	88	155	215	0.6
DOOR PHONE	127	99	32	0.2
DOOR PHONE INTERFACE MODULE (DPIM)	29	90	120	0.2
CTI INTERFACE MODULE	29	90	120	0.2

ENVIRONMENTAL LIMITS

OPERATING TEMPERATURE	0 - 40 °C
STORAGE TEMPERATURE	-10.5 - 70 °C

CABLE REQUIREMENTS

EQUIPMENT	CABLE	AWG	MAX FEET	MAX METRES
DIGITAL KEYSETS	1 PR. TWISTED	24	1300	400
SINGLE-LINE STATION	1 PR. TWISTED	24	3000	1 KM
DOOR PHONE	2 PR. TWISTED	24	330*	100
CTI INTERFACE MODULE	1 PR. TWISTED	24	1300	400

*This is the maximum length of the cable between the door phone and the DPIM. The DPIM can be installed up to 250 cable metres from the KSU.

Details of system tones may differ as specified by local regulations

SYSTEM TONES

TONE	FREQUENCIES	CADENCE
DIAL TONE	350 + 440 Hz	CONTINUOUS
RING BACK TONE	440 + 480 Hz	1 sec on + 3 sec off
BUSY TONE	480 + 620 Hz	0.5 sec on + 0.5 sec off
DND/NO MORE CALLS	480 + 620 Hz	0.25 sec on + 0.25 sec off
ERROR TONE	480 + 620 Hz	0.25 sec of each tone
CONFIRMATION TONE	350 + 440 Hz	Three bursts of tone: 0.1 sec on + 0.1 sec off
TRANSFER/CONF	350 + 440 Hz	0.1 sec on + 0.1 sec off

KEYSET LED INDICATIONS

CONDITION	LED COLOUR	LED ON	LED OFF
LINE IDLE	OFF -		OFF
LINE IN USE	RED/GREEN	STEADY	-
RECALL	AMBER	500 ms	500 ms
CALL ON HOLD	RED/GREEN	500 ms	500 ms
RINGING C.O. CALL	RED/GREEN	100 ms	100 ms
RINGING INTERNAL	CALL GREEN	100 ms	100 ms
DND INDICATION	RED 100 ms	ON/100 ms OFF for 500 ms	500 ms

RESERVE POWER DURATION ESTIMATES (minutes)*

	UPS CAPACITY IN VOLT AMPS (VA)						
NO. OF STATIONS	250	400	450	600	900	1250	2000

*These are approximate values based on an idle system. The greater the C.O. line activity on the system, the lower these readings will become. In addition, specific UPS devices, due to their internal construction, can have greater or lesser values.

PART 3. STATION EQUIPMENT

3.1 LCD 24B Model Keyset (Figure 3-1)

- Built-in speakerphone
- 24 programmable soft keys (16 with tri-coloured LEDs)
- Eight fixed-function keys
- 32-character display (2 x 16) with three associated soft keys and a scroll key
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
- Available in dark grey

3.2 STD 24B Model Keyset (Figure 3-2)

- Built-in speakerphone
- 24 programmable soft keys (16 with tri-coloured LEDs)
- Eight fixed-function keys
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
- Available in dark grey

Figure 3-1   **Figure 3-2**

3.3 LCD 12B Model Keyset (Figure 3-3)

- Built-in speakerphone
- 12 programmable soft keys (6 with tri-coloured LEDs)
- Eight fixed-function keys
- 32-character display (2 x 16) with three associated soft keys and a scroll key
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
- Available in dark grey

3.4 STD 12B Model Keyset (Figure 3-4)

- Built-in speakerphone
- 12 programmable soft keys (6 with tri-coloured LEDs)
- Eight fixed-function keys
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
- Available in dark grey

Figure 3-3   **Figure 3-4**

3.5 Enhanced 6B Model Keyset (Figure 3-5)

- Built-in speakerphone
- 6 programmable keys
- Four fixed-function keys
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
- Available in dark grey

3.6 Standard Model Keyset (DS-2100B) (Figure 3-6)

- Built-in speakerphone
- Four fixed-function keys
- UP/DOWN keys for digital control of speaker, handset and ringer volumes
- Eight selectable ring tones per keyset
- Desk-mounted or wall-mounted
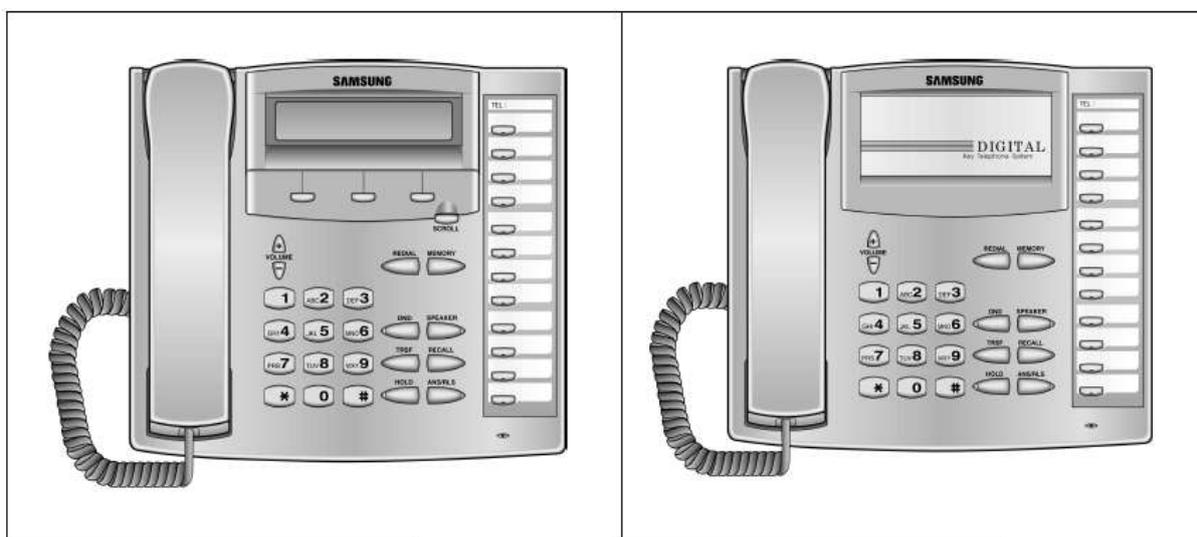- Available in dark grey
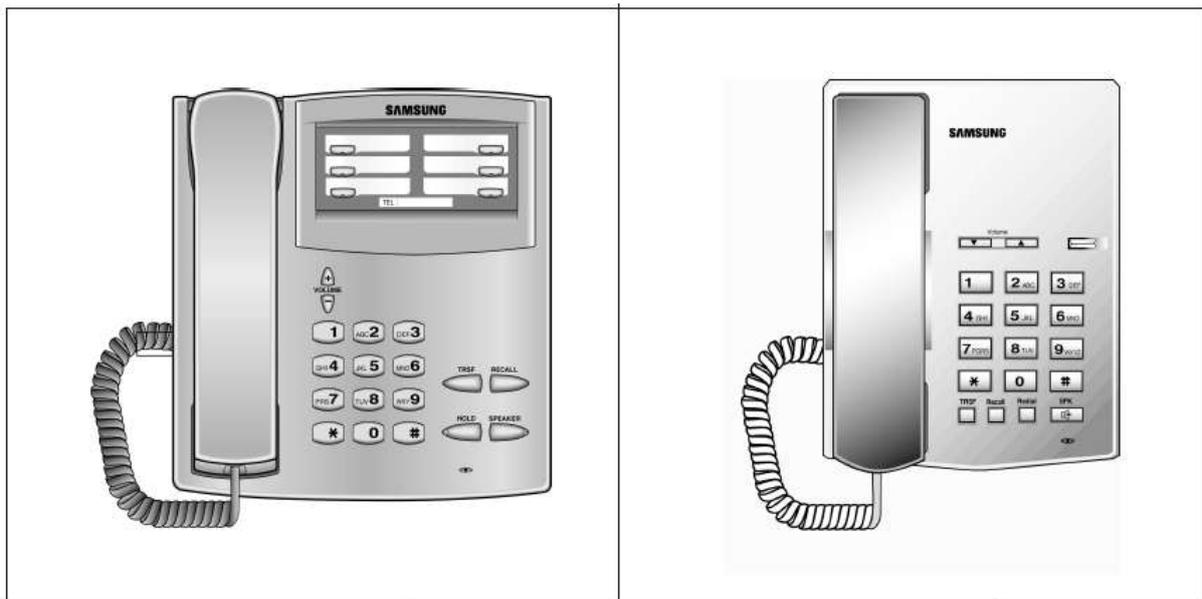
Figure 3-5

Figure 3-6

3.7 Door Phone Interface Module (DPIM) and Door Phone (Figure 3-7)

- The DPIM adapts any DLI circuit for use with the door phone unit
- Commonly used to request entry through locked doors (interior or exterior) or as a room monitoring box
- Door phone is wall-mounted
- Door phone is weather-proof

3.8 CTI Interface Module (CTM-TAPI) (Figure 3-8)

- Provides RS232 connection via DB9 for CTI
- Connects to any DLI circuit

Figure 3-7   **Figure 3-8**

PART 4. FEATURES

4.1 SYSTEM FEATURES

Account Code Entry
   Forced
   Voluntary
All Call Voice Page
Attention Tone
Authorisation Codes
   Forced
   Voluntary
Auto Attendant
   Automatic Hold
Background Music
Branch Groups
Call Forwarding
   All Calls
   Busy
   No Answer
   Busy/No Answer
   Follow Me
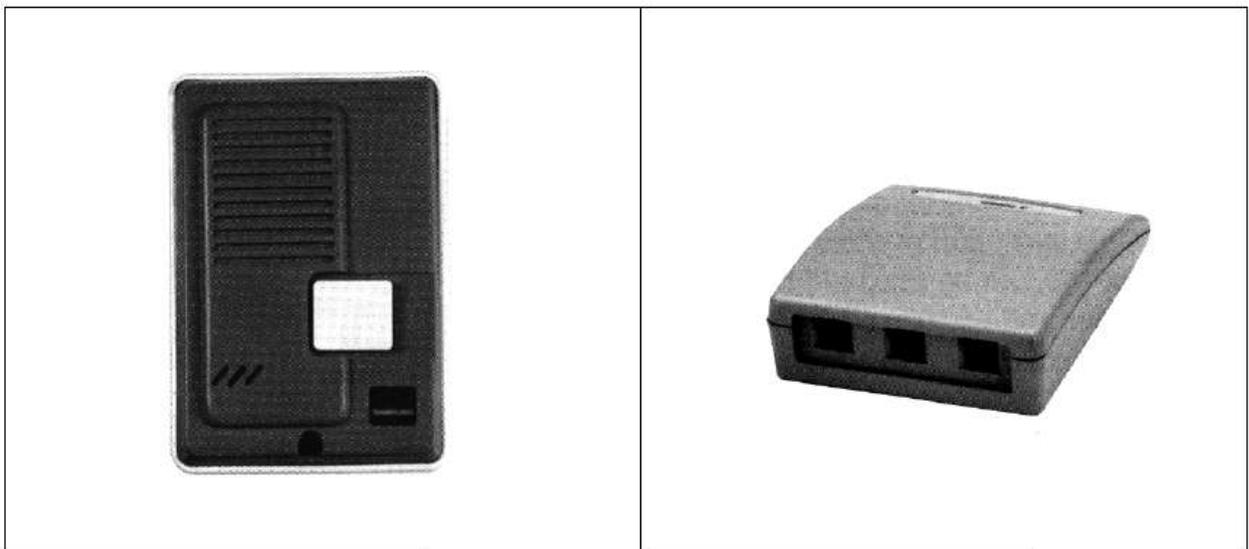   External
   To Voice Mail
Call Hold (Exclusive)
Call Hold (System)
Call Park and Page
Call Pickup
   Directed
   Groups
Call Waiting/Camp On
Chain Dialling
Class of Service
CLIP (Calling Line Identification Presentation)
   Name/Number Display
   Next Call
   Save CLIP Number
   Store CLIP Number
   Inquire Park/Hold
   CLIP Review List
   Investigate
   Abandon Call List
   CLIP On SMDR
   Number To Name Translation
Common Bell Control
Computer Telephony Integration (CTI)
TAPI
Conference
   Add On (5 parties)
   Unsupervised
Data Security
Direct Dialling Inward (DDI)
Direct In Lines
Direct Inward System Access (DISA)
Direct Trunk Selection
Directory Names
DISA Security
Distinctive Ringing
Door Lock Release (Programmable)
Door Phones
Door Phone Night Ring
Executive Barge–In (Override)
   With Warning Tone
   Without Warning Tone
Executive/Secretary Pooling

External Music Interfaces
External Page Interfaces
Flash Key Operation
Flexible Numbering
Flexible Ringing
Hot Line
In Group/Out of Group
Incoming Call Distribution
Incoming/Outgoing Service
Individual Line Control
ISDN service
   2BRI
   4BRI
Least Cost Routing
Live System Programming
Long Line Extensions
Meet Me Page and Answer
Memory Protection
Message Waiting Indications
Microphone On/Off Per Station
Music On Hold–Flexible
Night Service
   Automatic
   Manual
   Night Group
   Holiday
Operator Group
Orbit Parking
Overflow
   Operator
   Station Group
Paging
Power Failure Transfer
Prime Line Selection
Private Lines
Programmable Line Privacy
Programmable Timers
Recalls
Remote Programming–PC
Ring Over Page
Single Line Connections
Speed Dial Numbers
Speed Dial By Directory
Station Hunt Groups
System Alarms
System Directory
Station Message Detail Recording (SMDR) (Call Logging)

Toll Restriction (Call Barring)
Toll Restriction (Call Barring) Override
Tone or Pulse Dialling
Transfer
Trunk Groups
Uniform Call Distribution (UCD/ACD)
   Maximum of three Groups
   Call Statistics
   Agent Statistics
   Generic UCD/ACD Statistics
   Group Supervisors
Universal Answer
Voice Mail Integration
VPN
Walking Class of Service (WCOS)

4.2 SYSTEM FEATURE DESCRIPTIONS

Account Code Entry

Station users may enter an account code (maximum 12 digits) before hanging up from a call. This account code will appear in the last column of the SMD printout for that call record. Keyset users may enter this code using an account (ACCT) key without interrupting a conversation. Single line telephone users must temporarily interrupt the call by hook–flashing and dialling the feature access code. Account codes can be up to 12 digits long.

✆ **Forced**

When forced, account codes are always verified from a system list of 200 entries. Account codes are always printed on the SMDR report. They can contain digits 0–9.

✆ **Voluntary**

Users may select to enter an account code for any call. Digits can include 0–9, star (*) and #.

All Call Voice Page

Users can page all the internal and all the external paging at the same time by dialling the All Page code. Keysets may be restricted from making or receiving pages in system programming.

Attention Tone

To get your attention, a brief tone precedes all page announcements or intercom voice calls. There are separate programmable duration timers for page and voice announce tones.

Authorisation Codes

Authorisation codes are used to give permission to make a call. These four digit authorisation codes can be either forced or voluntary. When used, authorisation codes will automatically change the dialling station's class of service to the level assigned to the authorisation code. Authorisation codes may or may not be programmed to print on SMDR.

✆ **Forced**

When a station is programmed for forced authorisation the user must always enter this code before dialling is allowed. The dialled authorisation code is verified from a system list of 30 entries.

✆ **Voluntary**

Any station user can always enter an authorisation code before they begin dialling. The dialled authorisation code is verified from a system list of 30 entries.

Auto Attendant

The integrated digital automated attendant option provides four ports per AA option card. One AA option card per system. Sixteen professionally recorded prompts inform callers of the progress of their calls. Some examples are: "I am sorry. There is no answer", "That station is busy" and "Invalid number. Please try again". Two minutes

of battery–backed random access memory (RAM) provide up to 48 customer recordings for announcements or greetings. Twelve individual greeting boxes, each with its own dialling options, allow you to build call routing branches as needed. Callers are routed through the branches by dialling extension numbers or single digits.

NOTE
Requires optional hardware and/or software. Ask your dealer for details.

Automatic Hold

While a keyset user is engaged in an outside (C.O.) call, pressing another trunk key, route key or CALL key automatically places the call on hold when this feature is enabled. Pressing TRSF, CONFERENCE, PAGE or a DSS key always automatically places a C.O. call on hold. Intercom calls can be automatically held only by pressing TRSF or CONFERENCE. Each keyset user can enable or disable Automatic Hold.

Background Music

Keyset users may choose to hear music through their keyset speakers. Each user may adjust this level by the use of a volume control program at the selected keyset.

Branch Groups

Each station can assign its branch group. There are 8 branch groups in DCS- 816. When incoming or intercom call is ringing on station, any other stations that included the same branch group can answer by lifting the handset or pressing the speaker or ANS/RLS key.

Call Forwarding

This feature allows the user to redirect (forward) incoming calls. The calls can be redirected to the attendant, a hunt group, voice mail, external number or another station user. If the destination station is in Do Not Disturb (DND) mode, the calling party will receive DND/NO MORE tone. Calls cannot be forwarded to a door phone.

🕒 **All Calls**
This type of forwarding is not affected by the condition of the station. All calls are immediately redirected to the designated destination. If desired, the destination station may redirect the call back to the forwarded station by using the transfer feature. The forwarded station user can continue to originate calls as usual. If no key is programmed as Forward All, the TRSF key lights steady when a forwarded all condition is set.

🕒 **Busy**
This feature forwards all calls only when the station set is busy. The station user can originate calls as usual.

🕒 **No Answer**
This feature forwards calls that are not answered within a pre-programmed time. The user can originate calls as usual and receive calls if present. The timer is programmable on a per–station basis to allow for differences in individual work habits.

🕐 **Busy/No Answer**

This feature allows the station user to use both types of forwarding simultaneously, provided the destinations have already been entered in the usual manner.

🕐 **Follow Me**

This feature allows the station user to forward all calls from another station to the user's station or change the forward destination to the user's current location.

🕐 **External**

This feature forwards C.O. calls to an external number via a central office trunk if allowed by class of service. These C.O. calls forward only after the programmable external call forward delay timer expires.

🕐 **To Voice Mail**

Each station may be programmed to allow or deny the ability to forward intercom calls to voice mail.

Call Hold (Exclusive)

Outside calls can be placed on exclusive hold at any keyset by pressing the HOLD key twice during a call. Calls placed on exclusive hold can only be retrieved at the keyset that placed the call on hold. Intercom calls are always placed on exclusive hold.

Call Hold (System)

Outside calls can be placed on system hold at any station. Users may dial the access code or press the HOLD key. Calls on system hold may be retrieved at any station.

Call Park and Page

Each C.O. line has its own park zone. This simple method eliminates confusion and ensures that a park zone is always available. Pressing the PAGE key parks the call automatically. There are no extra keys to press and there is no lost time looking for a free zone.

Call Pickup

- **Directed**
  With directed call pickup, users can answer calls ringing at any station by dialling a code plus that station's extension number/incoming trunk number or by pressing the feature key and then dialling the extension number/incoming trunk number. You can pick up not only incoming and intercom call, but also hold recalling call if allowed.

- **Groups (8)**
  In addition, calls can be picked up from a station group in a similar manner. The group pickup feature allows users to answer any call ringing within any pickup group. There are 8 pickup groups available. A station cannot be in more than one pickup group. To use this feature, station users either dial the access code

or press the assigned feature key followed by the pickup group number. You can pick up not only incoming and intercom call, but also hold recalling call if allowed.

Call Waiting/Camp–On

Busy stations are notified that a call is waiting (camped–on) when they receive a tone. The tone will be repeated at a programmable interval. Keysets receive an off–hook ring signal through the speaker and single line stations receive a tone in the handset. The volume of the camp–on tone can be set by the station user. Camped–on calls will follow Forward No Answer if a Forward No Answer destination has been set.

Chain Dialling

Station users may manually dial additional digits following a speed dial call or chain together as many speed dial numbers as are required.

Class of Service

The system allows a maximum of 10 station classes of service. Each class of service can be customised in memory to allow or deny access to features and to define a station's dialling class. Each station can be assigned different classes of service for day and night operation.

CLIP (Calling Line Identification Presentation)

The CLIP feature is dependent on having an LCD keyset to show the name or number in the top line of the display.

- **Name/Number Display**
  Each LCD keyset user can decide if he/she wants to see the CLIP name or CLIP number in the display. Regardless of which one is selected to be seen first, the NND key is pressed to view the other pieces of CLI information.

- **Next Call**
  In the event that you have a call waiting or a camped–on call at your keyset, you can press the NEXT key to display the CLIP information associated with this next call in queue at your station. Either the CLIP name or CLIP number will show in the display depending on your NND selection.

- **Save CLIP Number**
  At any time during an incoming call that provides CLIP information, you may press the SAVE key. This saves the CLIP number in the Save Number feature. Pressing the SAVE number redial key will dial the CLIP number.

- **Store CLIP Number**
  At any time during an incoming call that provides CLIP information, you may press the STORE key. This save the CLIP numbers as a speed dial number in your personal speed dial list.

- **Inquire Park/Hold**
  Having been informed that an incoming call is on hold or has been parked, you may view the CLIP information before you retrieve the call. This will influence how you choose to handle the call.

- **CLIP Review List**
  This feature allows display keyset users to review CLIP information of calls sent to their stations. This list can be from 10 to 50 calls on a first–in, first– out basis. The list includes calls that you answered and calls that rang your station but that you did not answer. When reviewing this list, you can press one key to dial the person back.

- **Investigate**
  This feature allows selected stations with a special class of service to investigate any call in progress. If CLIP information is available for an incoming call, you will know to whom this station is speaking. On outgoing calls, you can see who was called. After investigating, you may barge–in on the conversation, disconnect the call or hang–up.

- **Abandon Call List (100)**
  The system has a system–wide abandon call list that stores CLIP information for the last 100 calls that rang but were not answered. The list is accessed using the operator's passcode. While reviewing this list, you are provided with options to CLEAR the entry or DIAL the number. You can use the NND key to toggle between the CLIP name, CLIP number and the date and time the call came in.

- **CLIP On SMDR**
  The Calling Line Identity of the calling party for each call, and the calling party name from the Translation table (if one has been entered by the user), will be printed on the SMDR record of each call.

- **Number To Name Translation (200)**
  The system provides a translation table for 200 entries. When the CLIP number is received, the table is searched. When a match is found, the system will display the corresponding name. This allows users in areas that do not support deluxe CLIP to provide names for regular callers.

Common Bell Control

DCS-816 provides a dry contact to control a customer–provided common bell or common audible device. This contact must be programmed as a member of a station group and may provide steady or interrupted closure.

Computer Telephony Integration (CTI)

Computer Telephony Integration (CTI) allows integration between the DCS-816 and a personal computer system (PC). Caller ID service is required on TAPI inbound call applications that use the CLIP information to display computer records in conjunction with the presentation of the call to the station on the DCS-816 system.

TAPI is described below.

🕛 **TAPI**

Jointly developed by Intel and Microsoft, TAPI (Telephony Applications Programming Interface) delivers telephony features to the Windows desktop. TAPI is an open application interface (OAI) protocol that supports First Party Call Control. A DCS-816 Computer Telephony Module (CTM) is required to connect any keyset to a PC running Windows 3.1 or Windows 95. The number of CTMs that can be installed on a system is only limited to the number of keyphones. The features and functionality of the DCS- 816 keyset are not changed.

Conference

The system allows five simultaneous conferences.

- **Add–On (5 PARTIES)**
  Any combination of up to five parties (stations or outside lines) can be joined together in an add–on conference. Parties may be eliminated or added after a conference has been established.

- **Unsupervised**
  A station user may set up a conference with two or more outside lines and then exit the conference leaving the outside lines connected in an unsupervised (trunk to trunk) conference.

Data Security

Single line extensions used with modems and facsimile machines can be programmed so that they will not receive any system–generated tones that would disrupt data transmissions. In addition, these devices will receive DCS- 816 C.O. ringing pattern instead of intercom ring pattern. Devices connected to an SLI port will receive a disconnect signal upon termination.

Direct Dialling Inward (DDI)

The system can use local telephone-company–provided DDI service via an ISDN. When programmed, anyone dialling a user's personal number will ring directly to that user's office. DDI calls to a busy station have the option to return busy signal to the C.O. or return ring-back to the C.O. When ring-back is selected, the called station receives off–hook ring. Multiple DDI numbers can ring the same extension or station group, and display keysets show a DDI directory name when ringing if a name has been programmed. DDI calls that are directed to ring a voice mail machine can be identified by a special digit.

Direct In Lines

Outside lines may be programmed to bypass the operator(s) and ring directly at any station or group of stations.

Direct Inward System Access (DISA)

Users can call in on specific DISA lines at any time, input a security code and receive system dial tone. Users can now place internal calls or, if permitted, calls using C.O. lines. The caller must have a tone dial phone and must know his/her DISA security code. DISA lines can be used as both–way lines or incoming only and may be active in day mode, night mode or both. The C.O. lines used for DISA must have disconnect supervision.

Direct Trunk Selection

Each station can be allowed access to or denied access from a trunk or trunk group by access code.

Directory Names

Each station, station group and C.O. line may be assigned a directory name (maximum 11 characters). In addition, each personal speed dial number, system speed dial number and entry in the DDI translation table may be assigned a name (maximum 11 characters). These names are displayed during calls with these ports and, in the case of station and speed dial names, can be used to originate calls. See the Dial by Name feature (in Station Features).

DISA Security

Telephone fraud and long–distance theft continue to increase; therefore, we have introduced a DISA security system. If an incorrect DISA passcode is entered repeatedly (as is the case with "hackers"), the DISA system can be automatically disabled temporarily. Both the number of incorrect passcode attempts and the time that DISA is disabled are programmable. In addition, all failed attempts to access DISA will print on SMDR (if provided) with a "DE" DISA error flag.

Distinctive Ringing

Users will know the type of call received by the type of ring heard. Outside calls have a single ring repeated while internal calls have a double ring repeated.

Door Lock Release (Programmable)

After answering a call from the door phone, users can dial a code to activate a contact closure. This can be used to operate a customer–provided electric door lock release mechanism. The contact closure timer is programmable from 100 to 2500 ms.

Door Phones

The door phone interface module (DPIM) provides for connection of a door phone to a DLI port. Pressing the key on the door phone produces a distinctive ring (three short rings repeated) at the assigned station or station group. If not answered within a programmable time, the system will release the door phone and stop the ringing. Stations may call the door phone directly and monitor the surrounding areas.

Door Phone Night Ring

The ring destination of door phone calls may be different at night than during the day. For example, large factories may want these calls directed to a security desk after hours.

Executive Barge–In (Override)

The feature allows specially programmed stations with a barge–in key to override the automatic privacy of another station. Programming allows barge–in with or without a warning tone. Stations may also be programmed as "secure" so that they cannot be barged–in on.

- **With Warning Tone**
  When the barge–in with tone option is set, the barging–in keyset has its microphone on and the barged–in on station receives an override display.

A double burst of warning tone sounds and repeats for every 10 seconds. This feature does not work from single line sets.

- **Without Warning Tone**
  When the barge–in without tone option is set, the barging–in keyset has its microphone muted and the barged–in on station does not receive an override display. This feature does not work from single line sets.

> **WARNING**
> BARGE–IN WITHOUT TONE MAY VIOLATE THE LAW CONCERNING THE RIGHT TO PRIVACY. SAMSUNG ELECTRONICS CO. LTD IS IN NO WAY RESPONSIBLE FOR THE POSSIBLE MISUSE OF THIS FEATURE.

Executive/Secretary Pooling

Each keyset may be defined as a BOSS or a SECRETARY in system programming. Each BOSS can have up to four SECRETARIES and each SECRETARY can have up to four BOSSES. These arrangements are known as executive/secretary pools. There can be multiple pools in a system. When a BOSS is in DND, all calls to the BOSS ring the first SECRETARY assigned to that BOSS; if that SECRETARY is busy, the call passes to the next available SECRETARY assigned to that BOSS. If the SECRETARY must communicate with the BOSS while he/she is in DND, pressing the corresponding BOSS key on the SECRETARY's keyset results in an Auto Answer intercom call being made to the BOSS (provided the BOSS should be free). A station can only be the BOSS of one SECRETARY pool. In addition, a station cannot be in more than one pool.

External Music Interfaces

DCS-816 provides an interface for connecting a customer–provided external music source. This source can be used for background music, station music on hold or trunk music on hold.

External Page Interfaces

DCS-816 provides one external page output and it can be assigned to operate with the dry contact in the system.

Flash Key Operation

While a user is on an outside line, pressing the FLASH key will flash the central office or PBX. This is used for custom calling features on C.O. lines or in conjunction with CENTREX/PBX operation. System programming allows individual flash times for C.O. and PBX lines. When C.O. or PBX flash is not required, setting the timers for two seconds releases the existing call and returns dial tone to make a new call.

Flexible Numbering

System programming allows stations to have two-, three- or four-digit extension numbers beginning with the digit 2. Default extension numbers begin with 201. station hunt group access codes can be two or three digits beginning with the digit 5. These can be changed but will affect other feature access codes. All user guides are written using the default numbering plan.

Flexible Ringing

Each C.O. line can be programmed to ring at any station or station group. Each line can be assigned a day ring destination and a night ring destination.

Hot Line

Stations can be programmed to call a pre–defined station or station group whenever that station goes off–hook. A hot line delay timer of 1–250 seconds can be programmed to allow sufficient time to make a different call.

In Group/Out of Group

Individuals assigned to a station hunt group may temporarily remove their telephones from the group by pressing the In/Out of Group key provided that there should be someone still in the group. Stations out of a group will not receive calls to that group but will continue to receive calls to their individual extension numbers. When desired, the user may put him/herself back into the group by pressing the key again. Users who do not have this key may dial the access code and the group desired. A station user is allowed to be in several groups if a key and the extender of that group are assigned for each group on the user's phone.

Incoming Call Distribution

Incoming calls can be assigned to ring a distributed station hunt group. This allows all members of the group to share the call load.

Incoming/Outgoing Service

Outside lines are available for incoming or outgoing service. Programming allows any outside line to be used for incoming calls only, outgoing calls only or both–way service.

Individual Line Control

Each station in the system can be individually programmed to allow or deny dialling out as well as allow or deny answering for each outside line.

ISDN Service

For DCS-816, the DCS Compact 2BRI or 4BRI card can be installed.

- **2BRI**   Provides 2 BRI ports, has four channels.

- **4BRI**   Provides 4 BRI ports, has eight channels.

Least Cost Routing

Least Cost Routing (LCR) is the ability to automatically select the appropriate central office route for the outside number dialled by any station. LCR can become quite complicated to understand and program but does allow highly complex dialling decisions. The LCR package includes the following features:

- Option to use or not use LCR
- Programmable LCR access code
- Digit analysis table: 300 entries each with 10 digits

- Routing by time of day and day of week
- Routing according to individual station class
- Modify digits table: 100 entries
- Flexible trunk group advance timer
- Option to use or not use trunk group advance warning tones
- Clear digits when dialling

Live System Programming

The system can be programmed from any display keyset or personal computer (PC) without interrupting normal system operation. There are three levels of programming: technician, customer and station. The technician level has access to all programs and can allow the customer access to system programs as needed. Technician and customer accesses are controlled by different security passcodes. Programming from a PC requires the PCMMC program.

Long Line Extensions

A single line (tip and ring) extension from the first two SLI ports may be connected to telephone company–provided long line circuits to remote locations.

Meet Me Page and Answer

After a user makes a Meet Me Page, the user may remain off–hook to allow the paged party to meet the user for a private conversation.

Memory Protection

In the event that power is lost to the system, all customer data contained in memory is retained for approximately seven days. In addition, the PCMMC computer program may be used to produce a backup copy of the customer data.

Message Waiting Indications

When calling a station and receiving a busy signal or the no answer condition, the caller can leave an indication that a message is waiting. The message key will flash red at the messaged keyset. A single line phone will receive a distinctive message waiting dial tone. Five message–waiting indications can be left at any station.

Microphone On/Off per Station

The microphone can be disabled at any keyset. When a microphone is disabled, the keyset cannot use the speakerphone, although on–hook dialling and group listening is still possible.

Music On Hold–Flexible

Each C.O. line may be programmed to receive chimed music source, external sources, internally–generated tones or no music when it is placed on hold. The system–generated tone is a beep for every 10 seconds.

Night Service

The system provides separate ringing locations for all trunks in both the day and the night modes.

- **Automatic**
  Automatic night service allows the system to automatically go in and out of night service according to the system clock. There are separate time options available for each day of the week. Auto night time is operated for trunks that have no night group (see below).

- **Manual**
  The operator presses the NIGHT key and then dials a passcode to change day mode to night mode operation.

- **Night Group**
  There are nine (1~9) night groups in DCS-816. Each trunk can have its own NIGHT group. When night start time is different for each office or trunk, you can change the DAY/NIGHT mode per NIGHT group by using the NIGHT key and night group number. If you use * as a group number, all trunks with or without a night group are changed to night mode simultaneously. This feature is operated manually only and is not affected by auto night time.

- **Holiday**
  There are 60 holiday tables. If it is a holiday, DCS-816 operates on night mode.

Operator Group

The station groups can contain 16 stations to answer incoming calls. Calls to this group can be set for distributed, sequential or unconditional ringing. Operators can use the In/Out of Group feature to meet flexible operator requirements. Default operator group is 500, but this can be changed to 501~509.

Orbit Parking

Park Slots enable SLT users to park a call into a number. They can announce the number to another user and that person can dial the Park Slot number to retrieve the call. This feature is handy for those people on SLT.

Overflow

- **Operator**
  When calls ringing an operator group go unanswered, they can overflow to another destination after a programmed period of time. The operator group has its own timer. The overflow destination can be a station or station group.

- **Station Group**
  When calls ringing a station group go unanswered, they can overflow to another destination after a programmed period of time. Each station group has its own timer. The overflow destination can be a station or station group.

Paging

System software allows for the use of four internal and one external paging zones. Stations can page any individual zone, all internal zones or all zones simultaneously. Using system programming, each station may be allowed or denied the ability to make and/or receive page announcement to/from any zone or combination of zones.

Power Failure Transfer

If power fails, these relays can be used to reroute the first two C.O. lines on the base board to SLTs (stations 15 and 16). When power is restored to the system, the lines and stations return to normal operation and calls in progress will be disconnected.

Prime Line Selection

Any station can be programmed to select a specific line, line group, telephone number.

Private Lines

For private line use, stations can be prevented from dialling and/or answering any line.

Programmable Line Privacy

Each outside line can be programmed to ignore the automatic line privacy. This allows up to four other parties to join your conversation by simply pressing the line key.

Programmable Timers

There are over 50 programmable system timers to allow each installation to be customised to best fit the end user's application.

Recalls

Calls put on hold, transferred or camped–on to any station will recall to the originating station if not answered within a programmable period of time. A recall that goes unanswered for the duration of the attendant recall timer will recall to the system operator group. Hold, transfer, camp–on and attendant recalls have individual programmable timers. Calls recalling to keys with tri–coloured LEDs will flash amber.

Remote Programming–PC

Remote programming allows the technician to access the system database from a remote location for the purpose of making changes to the customer data. Customer–provided modems, a PC using an optional software package and an SIO option card are needed to implement this feature.

Ring Over Page

Any outside line can be programmed to ring over a customer–provided paging system. Outside lines, door phones and station groups may ring over page in the day or night mode.

Single Line Connections

Single line ports allow for connection of a variety of SLTs plus facsimile machines, answering machines, loud bells, computer modems, cordless phones and credit card machines. When connecting customer–provided equipment to these extensions, compatibility should be checked before purchasing to ensure correct operation. Central office ring cadence can be selected for SLT stations. This is helpful when optional devices cannot detect intercom ring cadence.

Speed Dial Numbers

The system has a library of 500 speed dial numbers which may be allocated between the system list and stations as needed. The system list can have a maximum of 300 numbers and each station can have up to 50 numbers. Speed dial numbers are assigned in blocks of 10. Each speed dial number may contain up to 24 digits.

Speed Dial By Directory

The systems provide the user with the ability to look up a speed dial number and place the call. This feature can be used with the soft keys on the display keysets or as a programmable key.

Station Hunt Groups

System programming allows for up to 10 station hunt groups. One of three rings–sequential, distributed and unconditional–is available for each group. Each group may contain a maximum of 16 stations. A station may be assigned to more than one group. The default directory numbers to call these groups are 501–509. Each station group has its own recall timer for calls transferred to that group.

System Alarms

The alarm destination can be a station or station group. A DISA alarm warns the customer if the DISA security system has been triggered by too many incorrect password attempts. The alarm can ring any station or group of stations and show an appropriate display at the assigned stations.

System Directory

Each station, station group and outside line can have an 11–character directory name. This name appears on keyset displays to provide additional information about lines and stations.

Station Message Detail Recording (SMDR) (Call Logging)

The system provides, via an SIO option card, records of calls made, received and transferred. Connecting a customer–provided printer or call accounting system will allow collection of these records. Each call record provides the following details: station number, outside line number, start date, start time, duration of call, digits dialled (maximum 18), an account code (if entered) and call cost. (NOTE: Call cost is only available if relevant information is supplied by the network.) The system may print a header followed by 50 call records per page or send continuous records with no header for use with a call accounting machine.

The SMDR format contains many options that allow it to be customised for a company's individual needs. Options to print include incoming calls, outgoing calls, in and out of group status, change in DND status and authorisation codes.

Toll Restriction (Call Barring)

There are 200 allow and 200 deny entries of 11 digits each. Each of these entries can apply to dialling classes B, C, D, E, F and G. Expensive, premium rate and operator–assisted calls, as well as specific area and office codes, can be allowed or denied on a per–class basis. Class A stations have no dialling restrictions and Class H stations cannot make outside calls.

Any outside line may be programmed to follow station toll restriction or follow the toll restriction class assigned to it. Each station and trunk can have a day dialling class and a night dialling class.

Toll Restriction (Call Barring) Override

Program options allow system speed dial numbers to follow or bypass a station's toll restriction class. In addition, users may make calls from a toll restricted station using either the Walking Class of Service feature or the authorisation code feature.

Tone or Pulse Dialling

Outside lines can be programmed for either tone or pulse dialling to meet local telephone company requirements.

Transfer

System operation permits station users to transfer calls to other stations in the system. Transfers can be screened, unscreened or camped–on to a busy station.

Trunk Groups

Outside lines can be grouped for easy access by dialling a code or pressing a key. There are 4 trunk groups available. Access codes are 0 and 80–82.

Uniform Call Distribution (UCD/ACD)

UCD/ACD is used whenever the user expects to have more ringing calls than people to answer them. It prevents callers from receiving busy signals or lengthy delays before answering. Callers reaching a busy station group are held in queue for an available agent. First and second announcements reassure the caller until an agent becomes free. Up to three separate UCD/ACD groups can be created. Programmable automatic logout removes a station from the group if a call is placed to an unattended station, thus avoiding unanswered calls. A wrap–up timer prevents calls to a station for a programmable period of time to allow the agent to finish up work associated with the call.

- **Maximum of three Groups**
The UCD/ACD group option allows callers in queue at an UCD/ACD group to be diverted temporarily to a customer–provided announcement device and then placed back in the queue. A wrap–up timer allows agents to complete paperwork before receiving the next UCD/ACD call.

- **Call Statistics**
The UCD/ACD supervisor positions using a display keyset can monitor the number of calls in queue, the time that the oldest call has been waiting, the total number of calls received for the current day and the average time a call waits to be answered.

- **Agent Statistics**

  UCD/ACD supervisor positions using a display keyset can monitor the number of agents in a group and how many agents are currently logged in. Each station's status can be reviewed for the number of calls answered and the average call length of the current day.

- **Generic UCD/ACD Statistics**

  The following statistics report can be printed out on a per UCD/ACD group basis either once a day or on request by the supervisor.

Statistics	
Average Ring Time.....................................	0.09
Number of Times All Agents Busy....................	0
Average Time In Queue.....................................	0:00
Total Calls Received...	1
Longest Wait Time...	0:09

Agent	Calls RECEIVED	Average CALL TIME	Ring TIME
202	1	1:37	0:09
203	0	0:00	0:00
204	0	0:00	0:00

- **Group Supervisors**

  Multiple supervisors can be assigned to each group, or one station can be given supervisor status for multiple groups. The group supervisor (using a display keyset) can add and delete agents, in real time, to/from the group to handle the workload.

Universal Answer

Station users may dial the Universal Answer code or press the UA key to answer any outside lines programmed to ring the UA device. The UA device can be a station, group of stations, common bell or ring over page.

Voice Mail Integration

The system uses DTMF tones (inband signalling) to communicate with a third– party-supplied compatible voice mail system. Stations can call forward to a voice mail system. When answered, the system sends DTMF tones, routing the caller directly to the called station user's mailbox. Keyset users can press one key to retrieve messages from the voice mail system.

VPN

The VPN (Virtual Private Network) feature allows the DCS-816 to operate in a closed numbering group based on a 5 or 6 digit dialling plan for all members of the closed numbering group. In this case the local site DCS-816 extensions also operate with a same 5 or 6 digit extension number. The feature uses the LCR tables to identify and route calls within the closed numbering group and normal external calls. Local site calls are identified based on a programmable 2- digit self-identification code.

The system, when the feature is set to ON in MMC 210, automatically enters LCR mode and analyses the digits dialled. If the first two digits are the same as the system Self ID code programmed in MMC 724 the system will activate the feature based on

the following digits dialled, e.g. internal call. All other outgoing calls are routed in accordance with LCR related tables MMC 710, MMC 712 and MMC 713.

Walking Class of Service (WCOS)

This feature allows users to make calls or use features from a station that is restricted. The users may either use the WCOS feature code or the authorisation code feature. Both methods change the class of service corresponding to the station passcode or authorisation code that is dialled. After the call is completed, this station returns to its programmed class of service.

4.3 STATION FEATURES

Appointment Reminder	Off–Hook Ringing
Automatic Hold	Off–Hook Voice Announce (OHVA)
Automatic Privacy	One Time Do Not Disturb
Background Music	One–Touch Dialling Keys
Busy Station Callback	On–Hook Dialling
Busy Station Indications (BLF)	Programmable Keys
Call Forwarding	Programmed Station Messages
Call Pickup	Protection From Barge–In
Dial By Name	Pullout Directory Tray
Direct Station Selection (DSS)	Pulse To Tone Switch Over
DND Override	Redial
Do Not Disturb (Programmable)	Auto Retry
Door Lock Release	Last Number
Exclusive Hold	Save Number
Group Listening	Ring Modes
Headset Operation	Ringing Preference
Hearing Aid Compatible	Speakerphone
Line Queuing With Callback	Station Lock
Line Skipping	Tri–Coloured Lights
Loud Ringing Interface	Volume Settings
Message Waiting Light/Indication	Wall–Mountable Keysets
Mute Microphone/ Handset	

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4.4 STATION FEATURE DESCRIPTIONS

Appointment Reminder

Keysets with an alarm key can be used like an alarm clock. When programmed for a specific time, the keyset will sound a distinctive ring to remind you of meetings or appointments. Alarms can be set for "today only" or for every day at the same time. Up to two alarms may be set at each keyset. Display keysets can also show a programmed message when the alarm rings.

Automatic Hold

Station users can enable or disable automatic hold at their keysets. While a user is engaged on an outside (C.O.) call, pressing another trunk key, route key or

CALL key automatically puts the call on hold when this feature is enabled. Pressing TRSF, CONFERENCE, PAGE or a DSS key will always automatically place the call on hold. This type of automatic hold is not a user–selectable option.

Automatic Privacy

All conversations on outside lines and intercom calls are automatically private. The privacy feature can be turned off on a per–line basis.

Background Music

When a customer–provided music source is connected, each keyset user may listen to background music. The HOLD key turns background music on or off and the volume is controlled by the volume control keys. One music source can be externally connected to the system. Chime music is available by default.

Busy Station Callback

When reaching a busy station, callers may request a callback by pressing one key or dialling a code. The system rings the caller back when that station becomes idle (a system–wide maximum of 100 callbacks are allowed at one time, including busy station and busy trunk).

Busy Station Indications (BLF)

DSS/BLF keys may be assigned to any keyset. These keys will be off when the station is idle, light red when that station is in use and flash distinctively when that station is in DND mode.

Call Forwarding

Station users can forward internal and outside calls to other destinations immediately (**Forward All**), when busy (**Forward Busy**) or when not answered within programmable duration (**Forward No Answer**).

These forward destinations can all be different. Once a destination has been programmed, it can be turned on and off with a programmable key. Forward All Calls takes priority over Busy and No Answer conditions.

In addition to the three usual methods of forwarding described above, a fourth option called Follow Me is available. This option allows station users to set a Forward All condition from his/her station to another station while at the remote station. To display the Follow Me condition, the TRSF key lights steady red at the station that is forwarded. The TRSF key also lights if Forward All is set and no key is programmed for Forward All.

Keyset users can also be given an external call forward key to forward their calls to an external phone number. Each outside line may be programmed to either follow or ignore station call forwarding. A per–station option controls whether internal calls forward to voice mail or not. SLTs must have the system administrator program this feature for them.

Call Pickup

With directed call pickup, a user can answer calls ringing at any station by dialling a code plus that extension number. The group pickup feature allows the user to

answer any call ringing within a pickup group. Pickup keys may be customised with extenders to allow pickup from a specific station or pickup group. The DCS- 816 has 8 programmable pickup groups.

Dial By Name

Each system and personal speed dial number can have an associated directory name. A speed dial number can be selected by scrolling alphabetically through the directory name list. This on–line "directory" allows the user to look up and dial numbers in seconds.

Direct Station Selection (DSS)

Programmable keys can be assigned as DSS keys and associated with extension numbers. Users press these keys to call or transfer calls to the assigned stations.

DND Override

The Do Not Disturb (DND) feature is used to stop all calls to a station. However, an extension that has a class of service that allows DND override can override the DND status of another extension using the DNDOVRD soft key or DND OVERIDE ID.

Do Not Disturb (Programmable)

The Do Not Disturb (DND) feature is used to stop all calls to a station. System programming can allow or deny use of the DND feature for each station. Parties calling a station in DND will receive reorder tone. A keyset without a DND key can activate DND via the feature access code. The ANS/RLS key will flash at 112 ipm (rapidly) when DND is set. There is a programmable option to allow a C.O. line to override DND at its ring destination if that destination is a single station.

Door Lock Release

Stations programmed to receive calls from a door phone can dial a code to activate a contact closure for control of a customer–provided electronic door lock.

Exclusive Hold

Pressing the HOLD key twice will hold a call exclusively at a station so no other station can pick up that call. Intercom calls are automatically placed on exclusive hold.

Group Listening

This feature allows users to turn on the speaker while using the handset. This allows a group of people to listen to the distant party over the speaker without the microphone turned on.

Headset Operation

Every keyset can be programmed to allow for the use of a headset. In the headset mode, the hookswitch is disabled and the ANS/RLS key is used to answer calls. ANS/RLS key lights steady red when the keyset is in headset mode.

Headset operation can also be achieved using a third–party supplied headset adapter.

Hearing Aid Compatible

All keysets are hearing aid compatible.

Line Queuing with Callback

When the desired outside is busy, the user can press the CALLBACK key or dial the access code to place his/her station in a queue. The user will be called back when the line is available (a maximum of 100 callbacks are allowed system–wide at one time including busy station and busy trunk).

Line Skipping

When the user is talking on an outside line and the automatic hold feature is turned off, he/she may press an idle line key and skip to that line without causing the previous call to go on hold.

Loud Ringing Interface

Ring tone can be provided to the external page port connecting to a customer–provided amplifying device. The output can then be assigned to ring with a specific station to provide loud ringing capability.

Message Waiting Light/Indication

When a message indication is left at a keyset, the MESSAGE key will slowly flash red. SLTs will receive a distinctive dial tone to notify them that a message is waiting. Message waiting indications can be left for any station or group of stations.

Mute Microphone/Handset

Any keyset user can mute the keyset's handset transmitter by pressing the MUTE key. In addition, keyset users can also mute the keyset microphone while the keyset is in speakerphone mode.

Off–Hook Ringing

When a keyset is in use, the system will provide an off–hook ring signal to indicate that another call is waiting. The ring signal is a single ring repeated. The interval is controlled by a system–wide timer. Single line stations will receive a tone burst through the handset receiver instead of a ring.

Off–Hook Voice Announce (OHVA)

Keysets may receive a voice announcement while on another call. The calling station must have an OHVA key. When transferring a call to a busy keyset or while listening to busy signal, the station user can press the OHVA key to make an OHVA call to the busy keyset. If the called keyset is in the DND mode, it cannot receive OHVA calls.

One Time Do Not Disturb

The Do Not Disturb (One Time) feature is used to stop all calls to a station when the user is on an outside line and does not want to be disturbed for the duration of the call. Upon completion of the call, DND is cancelled and the station is returned to normal service. This feature requires a programmed key.

One–Touch Dialling Keys

Frequently–used speed numbers can be assigned to one–touch dialling keys for fast accurate dialling.

On–Hook Dialling

Any keyset user can originate calls without lifting the handset. When the called party answers, the user may speak into the microphone or lift the handset for more privacy.

Programmable Keys

LCD 24B and STD 24B keysets have 24 programmable keys; LCD 12B and STD 12B keysets have 12; Enhanced 6B keysets have 6. Each key can be programmed for different uses to personalise each phone. Examples of keys include individual outside line, individual station, group of lines, group of stations and one–touch speed dial keys. Using these keys eliminates dialling access codes.

The following feature keys have extenders that make them more specific: SPEED DIAL, SUPERVISOR, PAGE, DSS, DIRECTED PICKUP, GROUP PICKUP, DOOR PHONE, BOSS, PROGRAMMED MESSAGE, IN AND OUT OF GROUP and FORWARD. The extender can be a station, a group or another identifying number.

Programmed Station Messages

Any station may select one of 20 messages to be displayed at a calling party's keyset. Ten messages are factory–programmed and the remaining 10 can be customised by the system administrator (16 characters maximum).

NOTE
The calling party must have a display keyset to view these messages.

Protection From Barge–In

Each station can be programmed as secure or not secure. Secure stations cannot be barged–in on. A station that is not secure cannot be barged–in on when talking to a secure station.

Pullout Directory Tray

A pullout directory tray is conveniently located beneath all keysets. It is used to record station directory names and speed dial numbers.

Pulse To Tone Switch Over

When dialling a number on a dial pulse network, a station user can dial # and the system will begin to send DTMF.

Redial

There are three types of external redial available to all station users. Each type can redial up to a maximum of 18 digits.

- **Auto Retry** – When an outside number is dialled and a busy signal is received, the auto retry feature can be used to reserve the outside line and automatically redial the number for a programmable number of attempts.

- **Last Number** – The recently dialled most number on a C.O. line is saved and may be redialled by pressing the redial key or dialling the LNR access code.

- **Save Number** – Any number dialled on a C.O. line may be saved for redial at a later time.

Ring Modes

Each keyset user can select one of three distinct ways to receive intercom calls. The phone can automatically answer on the speakerphone, voice announce through the speaker or receive ringing. When the ring mode is selected, keyset users can choose one of eight distinct ring tones. Forced Auto Answer is invoked by the calling station and is controlled by the calling station's class of service.

Ringing Preference

Lifting the handset or pressing the ANS/RLS key automatically answers a call ringing at the keyset. Using this method, users are assured of answering the oldest call first. When ringing preference is turned off, the user must press the flashing key to answer. Users may answer ringing lines in any order by pressing
the flashing key.

Speakerphone

All keysets have a built–in speakerphone. This speakerphone enables calls to be made and received without the use of the handset.

Station Lock

With a programmable personal station passcode, any keyset can be locked and unlocked or only outside call. A locked keyset cannot be used to make or receive calls.

Tri–Coloured Lights

LCD 24B and STD 24B keysets have 16 keys equipped for tri–coloured LED indications (green, red and amber). LCD 12B and STD 12B models have eight of these keys. To avoid confusion, your calls always light green; other calls show red and recalls show amber.

Volume Settings

Each keyset may separately adjust the volume of the ringer, speaker, handset receiver, background music, page announcement and off–hook ring tone.

Wall–Mountable Keysets

Every keyset comes equipped with a reversible base wedge that can be utilised as a wall–mount bracket.

4.5 DISPLAY FEATURES

Account Code Display	Calling Party Number
Call Duration Timer	Conference Information
Call For Group Identification	Date And Time Display
Call Processing Information	Dialled Number
CLIP Information	Enhanced Station Programming
Name/Number Display	Identification of Recalls
Next Call	Identification of Transfers
Save CLIP Number	Message Waiting Caller Number
Store CLIP Number	Outside Line Identification
Inquire Park/Hold	Override Identification
CLIP Review List	Programmed Message Display
Investigate	Soft Keys
Abandon Call List(100)	Stopwatch Timer
Calling Party Name	UCD/ACD Supervisor Displays

4.6 DISPLAY FEATURE DESCRIPTIONS

Account Code Display

Account codes are conveniently displayed for easy confirmation. If entered incorrectly, users may press the ACCT key again and re-enter the account code.

Call Duration Timer

The system can automatically time outside calls and show the duration in minutes and seconds. Station users may manually time calls by pressing the TIMER key.

Call For Group Identification

When a call is made to a station group, the display shows [CALL FOR] and the user's group number. These calls can be answered with a different greeting than calls to the user's extension number.

Call Processing Information

During every-day call handling, the keyset display will provide information that is helpful and in some cases invaluable. Displays such as [CALL FROM 203], [TRANSFER TO 202], [701: RINGING], [TRANSFER FM 203], [708 busy], [Camp on to 204], [Recall from 204], [Call for 501], [message from 204] and [FWD ALL to 204] keep users informed of what is happening and where they are. In some cases, the user is prompted to take action and in other cases the user receives directory information.

CLIP Information

CLIP information is dependent on the use of LCD keysets. The following list explains the displays that are used with CLIP.

- **Name/Number Display**
  Each display keyset user can decide if he/she wants to see the CLIP name or CLIP number in the display. Regardless of which one is selected to be seen first, the N/N key is pressed to view the other piece of CLIP information.

- **Next Call**
  In the event that there is a call waiting or a camped–on call at your keyset, you can press the NEXT key to display the CLIP information associated with this next call in queue at the station. Either the CLIP name or CLIP number will show in the display depending on your N/N selection.

- **Save CLIP Number**
  At any time during an incoming call that provides CLIP information, you may press the SAVE key. This saves the CLIP number in the Save Number feature. Pressing the SAVE number redial key will dial the CLIP number.

- **Store CLIP Number**
  At any time during an incoming call that provides CLIP information, you may press the STORE key. This save the CLIP numbers as a speed dial number in the personal speed dial list.

- **Inquire Park/Hold**
  When a user is informed that an incoming call is on hold or has been parked, the user may view the CLIP information before he/she retrieves the call. This will influence how the user chooses to handle the call.

- **CLIP Review List**
  This feature allows display keyset users to review CLIP information for calls sent to their stations. This list can be in a first–in, first–out basis. The list includes calls that were answered and calls that rang the user's station but were not answered. When reviewing this list, the user can press one key to dial the person back.

- **Investigate**
  This feature allows selected stations with a special class of service to investigate any call in progress. If CLIP information is available for an incoming call, the selected stations can know to whom the DCS-816 user is speaking. On outgoing calls, the selected stations can see who was called. After investigating, the selected stations may barge–in on the conversation, disconnect the call or hang–up.

- **Abandon Call List**
  The DCS-816 has a system–wide abandon call list that stores CLIP information for the last 100 calls that rang but were not answered. The list is accessed using the operator's passcode. When reviewing this list, you are provided options to CLEAR the entry or DIAL the number. You can use the NND key to toggle between the CLIP name, CLIP number and the date and time the call came in.

Calling Party Name

For intercom calls, LCD 24B and LCD 12B keysets show the calling party's name before answering. The names must be stored in the system directory list and can be up to 11 characters long.

Calling Party Number

When an intercom call is received, all display stations show the calling party's extension number before the call is answered.

Conference Information

When a conference is set up, each extension and outside line number is displayed at the controlling station when it is added. When a station is added, its display shows [CONF with xxx] – alerting the user that other parties are on the line.

Date and Time Display

In the idle condition, the current date and time are conveniently displayed. Display keysets can have 12– or 24–hour clock in either ORIENTAL or WESTERN display format with information shown in uppercase or lowercase letters.

Dialled Number

When an outside call is made, digits are displayed as the user dials them. If the display indicates an incorrect number was dialled, the user can quickly hang up before billing begins.

Enhanced Station Programming

Personal programming options are easier to select and confirm with the help of the display.

Identification of Recalls

Hold recalls and transfer recalls are identified differently from other ringing calls. Hold recalls indicate the recalling line or station number and the associated name. Transfer recalls indicate the recalling line or station and where it is coming from.

Identification of Transfers

The display identifies who transferred a call to the user.

Message Waiting Caller Number

When the message indication is on, pressing the MESSAGE key displays the station number(s) of the person(s) who have messages for the user. Display keyset users can scroll up and down to view message indications.

Outside Line Identification

Each line can be identified with an 11–character name. Incoming calls display this name before the call is answered. This feature is helpful when individual lines must be answered with different greetings.

Override Identification

If another station barges–in on a user's conversation, the display alerts the user with a [Barge from 2xx] display if the system is set for barge–in with tone.

Programmed Message Display

Pre-programmed station messages set by other stations are displayed at the calling station's keyset.

Soft Keys

Below the display are three soft keys and a SCROLL key. These keys allow the user to access features in his/her class of service without requiring the keyset to have designated feature keys.

Stopwatch Timer

Display keyset users find this feature very convenient to time meetings, calls and other functions. Users simply press once to start the timer and press again to stop the timer.

UCD/ACD Supervisor Displays

With the optional AA card, when UCD/ACD is used, multiple supervisors can view useful statistics about the UCD/ACD group. Statistics are broken down into agent or calls. Agent statistics allow the supervisor to monitor and change the status (in group, out of group and DND) of any agent, or view each agent's total number of calls or average call length. Call statistics allow the supervisor to view how many calls are in queue, the longest wait time, how many calls have been received today and the average time in queue.

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