# Design Your Business Communications with iPECS UCP 


#### Abstract

iPECS UCP is Ericsson-LG Enterprise's Unified Communications Platform designed to meet SME and Enterprise communications needs. As ground breaking innovative platform, iPECS UCP provides out of box UC and Mobility solution. In addition, iPECS UCP is scalable for premium UC solution.


## Embedded UC and Telephony

As the most compelling advantage, UC features are embedded in iPECS UCP. Users can improve business efficiency and productivity with embedded UC features including real-time voice, video and presence enabled IM with messaging services (Visual Voice Mail and SMS etc.) under a single user interface on multiple devices.
Rich Business Applications
iPECS UCP provides a various range of applications and mobile clients to fulfill varying needs and requirements in the SME environments. Also, it offers interoperability with various 3rd party solutions in hospitality, healthcare and other vertical industries.
Wide Range of Mobility
iPECS UCP provides multiple mobility solutions to improve business productivity and decrease communication expenses. iPECS DECT and Wi-Fi phones provide feature rich and reliable communications for internal mobile workers. iPECS UCS mobile client delivers the power of a desk phones to smartphones or tablet PCs for external mobile workers.
Ultimate Flexibility on Deployment
iPECS UCP as "Modular ALL-IP Architecture" enables flexible and cost-effective multi-site deployment with industry-unique architecture. Furthermore, virtualized deployment is supported to maximize flexibility on deployment as well. (iPECS vUCP is needed) It can be fully networked in T-NET (Transparent Networking) or Networking mode for local survivability and PSTN failover. Also, it provides call server (Local/Geo) and power redundancy for seamless communications environment. Through flexible T-NET and redundancy features, IT managers can easily manage a multi-site architecture. Leverage your business to full of flexibility with wide range of deployment scenarios.
Seamless Scalability
As a scalable call server iPECS UCP allows businesses to easily expand capacity with optional gateways or boards. In addition, simple system port licensing enables upfront investment savings and cost-effective expansion as a business grows. Not only system capacity, communications features including UC solution also can be expanded or added with simple license.
Virtualize Your Business Communications
iPECS vUCP (virtual UCP) is the first iPECS Unified product for virtualization solution. iPECS vUCP, as a software base call server is running on VMware virtual environment. The software is same as on-premise iPECS UCP and all the system features as well. Under the equal user experiences, iPECS vUCP will make easy to deploy your business communication's flexibility and scalability. Also, it delivers an innovative way to manage diverse business applications and collaborate cost-effectively, so you can leverage business communications while keeping all the advantages of iPECS UCP.



## Key Features

- Built-in iPECS UCS
- Built-in iPECS ClickCall
- Embedded VoIP

Embedded Voice Mail
Embedded ACD

- Embedded SIP
- Embedded audio conference
- Embedded hotel features
- Mobile extension
- One number service
- Web call back
- ACD call statistics for multiple group
- Emergency/Alarm call service
- Simplified directory search and dial

Custom MOH support

- VM to E-Mail forwarding
- Centralized Voice Mail
- Centralized control T-NET
- System geographic redundancy
-Web administration
- Web user portal
- Virtualization with iPECS vUCP


## Applications

- iPECS UCS
- iPECS ClickCall
- iPECS RCC Gateway
- iPECS Attendant (Office/Hotel)
- iPECS IPCR
- iPECS CCS
- iPECS Report Plus
- iPECS NMS


## Supported Terminals

- LIP-9071
- LIP-9000 Series

LDP-9200 Series

- GDC-800H (IP DECT)
- GDC-500H/480H
- WIT-400HE


## $3^{\text {rd }}$ Party Applications and Middleware

- Microsoft Lync/SfB
- Fidelio I/F
- Tiger TMS for hospitality solution
- 3rd party PMS Interface
- LAS for healthcare solution
- TAPI (3rd Party)

System Capacity

| Category | UCP100 | UCP600 | UCP2400 | vUCP | Remark |
| :---: | :---: | :---: | :---: | :---: | :---: |
| Main cabinet |  | 10 Slot |  | - | $10^{\text {th }}$ slot is for PSU |
| System ports (Default) | 199 (50) | 600 (100) | 2,400 (600) | 2,400 (200) |  |
| IP Extension (Default) | 199 (30) | 600 (10) | 2,400 (10) | 2,400 (10) |  |
| CO/IP Line | 199 | 600 | 998 | - |  |
| Default Option 1 (Desktop/Mobile) | 2/2 | 10/10 | 20/20 | - | Only one default option |
| UCS Standard clients Default Option 2 (Advanced User) | 2 | 10 | 20 | 10 | can be used |
| Max | 100 | 200 | 400 | 400 |  |
| UCS Premium clients | 199 | 600 | 2,400 | 2,400 |  |
| Clickcall (Default) | 199 (2) | 600 (10) | 2,400 (10) | 2,400 (10) |  |
| Server Redundancy Local | - | 0 | 0 | - |  |
| Seographical | - | 0 | 0 | 0 | w/ License |
| Integrated Standard | 2FXS(SLT) | - | - | - |  |
| Telephony ports Option | 4 CO or 2BRI or 4BRI | - | - | - |  |
| Built-in Basic* | 2~6 | 6 | - | - |  |
| VoIP Channel $\quad$ Built-in Max** | 16 | 24 | - | 250 (per VVOIM) | w/ License(8 ch increment) |
| System Max | 199 | 600 | 998 | 998 (w/ vVOIM) | w/ VOIM and VCIM |
| Built-in Audio Conference | 6/10/14/18 | 6/18 | - | - | VoIP DSP channels can be assigned to the MCIU, 2 VoIP $=4$ Conf channels |
| Built-in Basic | $8 \mathrm{ch} / 4 \mathrm{hrs}$ | $8 \mathrm{ch} / 6 \mathrm{hrs}$ | - | - |  |
| Built-in Max | $8 \mathrm{ch} / 14 \mathrm{hrs}$ | $8 \mathrm{ch} / 16 \mathrm{hrs}$ | - | - | w/ License |
| VM Channel External Basic (UVM) |  | $8 \mathrm{ch} / 50 \mathrm{hrs}$ |  | - |  |
| External Max (UVM) |  | $16 \mathrm{ch} / 200 \mathrm{hrs}$ |  | $150 \mathrm{ch} / 500 \mathrm{hrs}$ (vUVM) | w/ License |
| UVM/vUVM per system |  | 30/0 |  | 30/1 |  |
| MCIM ${ }^{* * *}$ per system |  | 30 |  | 30 |  |
| Attendant |  | 50 |  | 50 |  |
| Serial Port (RS-232C) / USB (3.0) Host Port |  | 1/1 |  | - |  |

## Power Requirement

| Item | Description | Specification |
| :---: | :---: | :---: |
| Module | AC Voltage Input | $100 \sim 240 \mathrm{VAC}+/-10 \%$ @ $50 / 60 \mathrm{~Hz}$ |
| AC/DC Adapter | AC Current Input | 1.0 amps |
|  | DC Output | 48 VDC @ 0.8 amps |
| Keyset | AC Voltage Input | $100 \sim 240 \mathrm{VAC}+/-10 \%$ @ $50 / 60 \mathrm{~Hz}$ |
| AC/DC Adapter | AC Current Input | 0.2 amps |
|  | DC Output | 48 VDC @ 0.3 amps |
| PSU | AC Voltage Input | $100 \sim 240 \mathrm{VAC}+/-10 \%$ @ $50 / 60 \mathrm{~Hz}$ |
|  | Fuse | T6.3, AC250V |
|  | DC Output | $48 \mathrm{VDC}, 5.3 \mathrm{amps} / 5 \mathrm{VDC}$ @ 1 amps |

Dimension \& Weight

|  | Standard Gateway Module | $38.8 \times 230 \times 194.5$ |
| :---: | :---: | :---: |
| WxHxD <br> $\mathbf{( m m )}$ | Main Cabinet, Enhanced | $440 \times 265.6 \times 318.2$ |
|  | 19" Rack Mount modules | $436.6 \times 53 \times 318$ |
|  | Standard Gateway Module | 1.5 |
| Weight <br> $\mathbf{( K g )}$ | Main Cabinet, Enhanced <br> (with PSU module) | $7.78(9.32)$ |
|  | 19" Rack Mount modules | 4.32 |

## Operating Environment

| Temperature | $0\left({ }^{\circ} \mathrm{C}\right)-40\left({ }^{\circ} \mathrm{C}\right) / 32\left({ }^{\circ} \mathrm{F}\right)-104\left(^{\circ} \mathrm{F}\right)$ |
| :---: | :---: |
| Humidity | $0-80 \%($ Non-condensing $)$ |

## System Components

| Item | Board | Description |
| :---: | :---: | :---: |
| UCP Call Server | UCP100 | Unified Communications Platform Server 100, Basic 50, Up to 199 port |
|  | UCP600 | Unified Communications Platform Server 600, Basic 100, Up to 600 port |
|  | UCP2400 | Unified Communications Platform Server 2400, Basic 600, Up to 2,400 port |
|  | vUCP | Virtual Unified Communications Platform, Basic 200, Up to 2,400 port |
| Trunk Gateway | VCIM* | VOIP/Audio conference Module |
|  | VOIM8/24 | 8/24 ch VoIP Interface Module |
|  | LGCM4/8 | 4/8 port Analog CO Interface Module |
|  | BRIM2/4 | 2/4 BRI(4/8 ch) Interface Module |
|  | PRIM | 1 port PRI(30 ch) Interface Module |
|  | vVOIM | Virtual VoIP Interface Module |
| Extension Gateway | DTIM8/24 | 8/24 port Digital Line Telephone Interface Module |
|  | SLTM4/8/32 | 4/8/32 port Single Line Telephone Interface Module |
|  | WTIM4/8/24 | 4/8/24 port Wireless Telephone Interface Module (DECT base controller) |

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[^0]:    * 64 DSP channels by default and expandable up to 128 DSP channels

    VoIP and Audio Conference can be mixed use. 1 VoIP channel uses 2 DSP channels and 1 Audio Conference channel uses 1 DSP channel

