

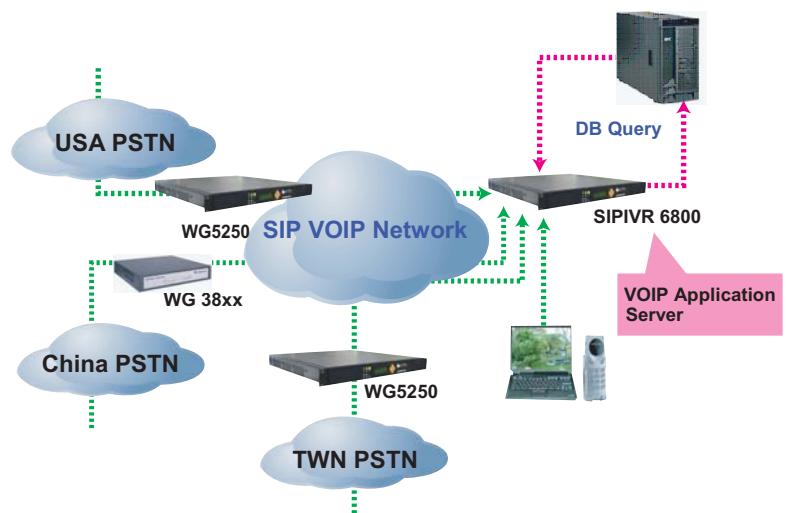
# SIPIVR 6800

## Service Creation Application Server

**SIPIVR 6800** brought to you a fully web user interfaced **Value Added Service Creation Application Server**. By using easy **drag and drop web interface**, you can create your owned VOIP application or value added service very quick time to market. With **built-in rich pre-designed** components, the developer can create their service without pay attention to the complexities of programming. Also the **real time debugger** makes developer very easy to debug and trace the call flow status.

### Selected Features

- Up-to 120 Universal VOIP Channels
- SIP RFC 3261 Compliance
- Fully Web Management Interface
- Audio Codec G.711, G.723.1, G.729A
- Drag and Drop Call Flow Editor
- Real Time Status/Variable debugger
- Rich-set of predefined components:
  - Basic flow components
  - IVR components
  - Database components
  - Flow control components
  - RADIUS components
  - Channel components
  - External Customized components
- Support Call Hold and Transfer
- Support in-band and out-of-band DTMF relay
- Support Database Connection Pools
- Support Job Push & Retrieve
- Free Text Expression
- Hitless Call Flow Update
- Optimized Developing Platform



### Application Examples

- Announcement Service
- Audio Broadcasting Service
- Auto Attendant
- Prepaid Calling Card Service
- Universal Call Back System
- IP Centrex Service
- Coloring Ring Back Tone Service
- Voice Mail System
- Interactive Voice Response Service
- Outbound Dialer System
- VOIP Call Center

