



FIBERME FCM5404 is an IP PBX appliance designed to bring enterprise-grade unified communications and security protection to all levels of businesses at an unprecedented price point without any licensing fees, costs-per-feature, or recurring fees. It enables enterprises to unify multiple communication technologies, such as comprehensive voice, fax, calling, conferencing, video/audio, data tools, security, mobility.

With an advanced hardware platform, software functionalities, FIBERME FCM5404 can support up to 300 registered users and offer effortless setup and deployment via the web-browser user interface. Besides auto-discovery of diverse endpoints and auto-provisioning, the FCM series IP-PBX offers a set of comprehensive features, including customizable call-routing, multi-level IVRs, call queues, auto-attendant, call detail records (CDR), multi-site peering, voicemail/fax forwarding to email.. And more

## Up to 300 Users

Supports up to 300 users and up to 60 concurrent calls

## CODEC Support

Supports Full-Band Opus voice codec, jitter resilience up to 50% packet loss

## Analog Ports

Built-in 4FXO for PSTN Lines and 2 FXS Ports for analog phones

## High Security

Highest level of security protection using SRTP, TLS and HTTPS encryption; Hi-speed network ports with Integrated NAT router and built-in firewall, support PIN code setting to customize dynamic firewall.

## High User-Friendliness

leverage autoclip intelligent inbound routing mechanism. With call records server(CDR), this IP-PBX can intelligently match inbound call number with historic called one in autoclip. Moveable extension, call forwarding, DND, Multi-language auto-attendant, LDAP and XML phonebooks, flexible dial plan

## Intelligent Resource Allocation

Optimize system resource utilization and system efficiency via stochastic algorithm, effectively minimizing hitting over processor resource and improving reliability in any scenarios.

## Excellent Compatibility

Without NAT traversal setting, this IP-PBX could be compliant with other mainstream SIP endpoints or components with changeable IP addresses, which effectively reduces complexity of configuration.

<b>Interfaces</b>	<ul style="list-style-type: none"> <li>• 2 Analog FXS Ports with lifeline capability in case of power outage</li> <li>• PSTN Line FXO Ports : 4 ports</li> <li>• Network Interfaces : Dual10/100 RJ45 ports</li> <li>• NAT Router</li> <li>• Peripheral Ports : USB, TF</li> <li>• LED Indicators :Power/Ready, Network, PSTN Line, USB, TF</li> <li>• Reset Switch</li> </ul>
<b>Voice/Video Capabilities</b>	<ul style="list-style-type: none"> <li>• Voice-over-Packet Capabilities: LEC with NLP Packetized Voice Protocol Unit, 32~128ms-tail-length carrier grade Line Echo Cancellation, Dynamic Jitter Buffer</li> <li>• Voice and Fax Codecs: G.711 A-law/U-law, G.722, G.723.1 5.3K/6.3K, G.726, G.729A/B, GSM, AAL2-G.726-32; T.38</li> <li>• QoS: Multiple Layers</li> </ul>
<b>Signaling &amp; Control</b>	<ul style="list-style-type: none"> <li>• Supported Dial Mode: DTMF and Pulse DTMF Methods: In Audio, RFC2833, and SIP INFO</li> <li>• Provisioning Protocol &amp; Plug-and-Play: TFTP/HTTP/HTTPS, auto-discovery &amp; auto-provisioning of various IP endpoints</li> </ul>
<b>Technical Add-on Features</b>	<ul style="list-style-type: none"> <li>• Caller ID :Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF</li> <li>• Polarity Reversal/Wink, with enable/disable option upon call establishment and termination</li> <li>• Maximum Call Capacity: Up to 50 even in SRTP encrypted</li> <li>• Call Features: Call park, call forward, call transfer, DND, ring/hunt group, paging/ intercom etc.</li> </ul>
<b>Network Protocols</b>	<ul style="list-style-type: none"> <li>• TCP/UDP/IP, RTP/RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP/HTTPS, PPPoE, SIP(RFC3261), STUN, SRTP, TLS, LADP</li> <li>• Disconnect Methods: Call Progress Tone, Polarity Reversal, Hook Flash Timing, Loop Current Disconnect, Busy Tone</li> </ul>
<b>Application Enhancements</b>	<ul style="list-style-type: none"> <li>• Call Center: Multiple configurable call queues, automatic call distribution (ACD) based on agent skills/availability/ busy level, in-queue announcement</li> <li>• Customizable Auto Attendant: Unlimited layers of IVR (Interactive Voice Response)</li> <li>• Conference Bridges: Up to 25 simultaneous PSTN or IP participants</li> </ul>
<b>Security</b>	<ul style="list-style-type: none"> <li>• Media Encryption: SRTP, TLS, HTTPS, TELNET with Fail2ban, Whitelist, Blacklist, alerts and more to protect against attacks</li> </ul>
<b>Multi-Language Support</b>	<ul style="list-style-type: none"> <li>• English/Chinese/Traditional Chinese for Web UI; Customizable IVR/voice prompts for English, Chinese, British English;</li> <li>• Customizable language pack to support any other languages</li> </ul>



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