## **SIP Session Timer Function Added**

Category	S/W Release Version	Date
SIP	8.00	Apr. 2004

## **Function Description**

The gateway periodically checks the SIP session connection state by sending INVITE messages. Without mechanism like this, the gateway may continue to retain call connection state in case of network problems because it does not know that the session has ended. The INVITE message ensures that active sessions are forcefully terminated when network fault is occurred via periodic monitoring.

This feature is optional marked at the **DRAFT**, not at the **RFC** standard. Thus this function is vendordependant. The gateway pre-determines the peer's capability during the call setup and it doesn't send the **Session Timer** to the remote peer without **Session Timer** capability.

The users are not allowed to "disable" this function





## **Related Commands**

(config)# sip-ua (config-sip-ua)# min-se <sec>

Default: 1800sec