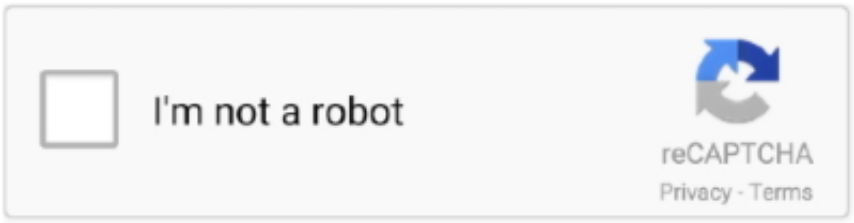

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VoIP Origination

The Radius server will then answer the gatekeeper with a yes or a no If a yes answer is received from the Radius server, the user the has the ability to make a phone call from the origination point.. How Call Origination WorksCall origination is a fairly simple process that is a bit more complicated to explain.. It's important to understand that a digital line can support a large amount of calls at one time, which can range anywhere from 20 to 30 depending on the type of line that is chosen.. It's important to note that a phone call through VoIP starts between the initiation point and the destination point, which are referred to as the originator and terminator respectively.. Since a gateway must interact with both the internet and standard PSTN lines, there are two interfaces necessary for a gateway, including a telephony interface that takes digital and analog lines and an Ethernet interface as a connection between the gateway and the internet.. What is Call Origination?VoIP stands for "Voice Over Internet Protocol" This means that phone calls utilize a technology that allows the calls to be sent directly over internet networks, which is a much cheaper way to make calls.. Once this is done, the user can make and receive calls fromhttp://www voip-info.. The provider for the PC to phone service will place a dialer on the internet that the user of the service can download and install.. It is at this point where the gatekeeper will interact with the Radius server again to see if the user in question has enough money to make the phone call.

There are typically different types of call origination depending on the services provided by the ITSP involved.. Once this has been determined, the relevant information will then be sent back to the Radius server.. In comparison to a digital line, an analog line can only support one phone call at a time.. This gateway will send the call to the final destination If no other issue arises during this time, the call can take place and the two users can talk for however long the duration of the call is.. This dialer allows for an account to be created with any type of payment source that is allowed by the service provider.. All that is necessary is a gateway, which essentially transfers phone calls from the internet and onto PSTN lines.. Once this has been determined, the relevant information will then be sent back to the Radius server.. If the call takes place once the maximum time allotment has been reached for the call, it will be disconnected and more money will need to be provided in order to make another call and continue the conversation.. When the call has come to a conclusion, whether by the callers themselves or because they were disconnected, the initial user will have the total price of the call deducted from their payment source by the billing server.. Required HardwareThe best aspect of VoIP services is that there are hardly any noteworthy hardware requirements.

voip origination vs termination

voip origination vs termination, voip origination provider, termination origination voip, origination services voip, sip termination vs origination, what is voip termination

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termination origination voip

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sip termination vs origination

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